

32-Bit 384-kHz Hi-Fi Audio Codec

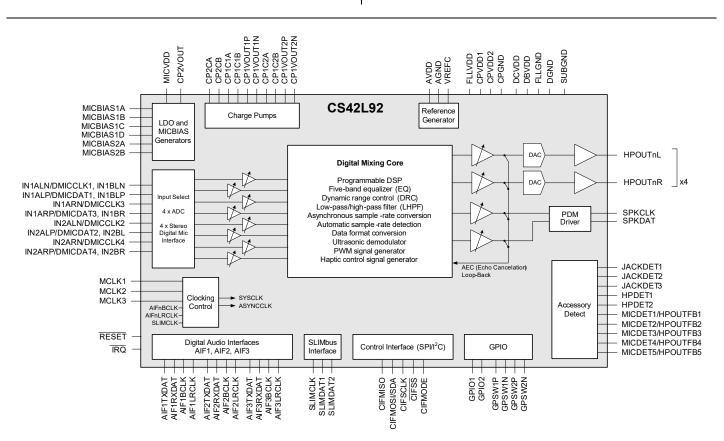
Features

- Integrated multichannel 32-bit hi-fi audio hub codec
- 99-dB signal-to-noise ratio (SNR) mic input (48 kHz)
- 127-dB SNR headphone playback (48 kHz)
- - 100-dB total harmonic distortion + noise (THD+N)
- Ultrasonic input- and output-path support
- Up to eight analog or digital microphone (DMIC) inputs
- Multipurpose headphone/earpiece/line output drivers
 - Support for balanced headphone output loads
 - 33 mW into 32- Ω load at 0.1% THD+N
 - Hi-fi filters for audiophile-quality playback
- Native audio playback up to 384 kHz sample rate, concurrent with voice and ultrasonic input signal paths
- Digital pulse-density modulation (PDM) output interface

- · Multichannel asynchronous sample-rate conversion
- Multiline SLIMbus[®] audio and control interface
- Three multichannel digital-audio interfaces
 - Standard data formats up to 384 kHz, 32 bits
- Flexible clocking, derived from MCLKn, AIFn, or SLIMbus
 - Low-power frequency-locked loops (FLLs) support reference clocks down to 32 kHz.
- · Advanced accessory detection functions
- Configurable functions on up to 16 general-purpose input/output (GPIO) pins
- · Integrated regulators and charge pumps
- Small W-CSP package, 0.4-mm staggered ball array

Applications

· Smartphones, tablets, and multimedia handsets







Description

The CS42L92 is a highly integrated low-power audio system for smartphones, tablets, and other portable audio devices. Multiple input/output paths are supported by a fully flexible all-digital mixing and routing engine, incorporating sample rate converters and other signal-processing functions for wide use-case flexibility. The integrated DSP provides a general-purpose signal processing capability; this is supported by general-purpose timer and event-logger functions.

The digital audio interfaces and hi-fi DACs enable 32-bit playback through the entire signal chain. Native audio playback at sample rates up to 384 kHz is possible, concurrent with voice and ultrasonic input paths.

The CS42L92 supports up to eight analog inputs and up to eight PDM digital inputs. Low-power input modes are available for always-on (e.g., voice-trigger) functionality using either analog or digital input. A smart accessory interface, with multipurpose impedance sensing and measurement capability, supports detection of external headsets and push buttons. Dual headphone connections (e.g., 3.5-mm and USB-C[™]) can be detected simultaneously.

Four hi-fi quality stereo headphone drivers are provided, each supporting stereo ground-referenced or mono bridge-tied load (BTL) configurations. Multiple headphone/earpiece outputs can be supported, including balanced stereo headphone configurations. The output drivers offer noise levels as low as 0.63 μ V_{RMS} into line or headphone loads. Selectable hi-fi filters support playback modes at sample rates up to 384 kHz.

Two channels of PDM output (one stereo interface) are available, and also an IEC-60958-3–compatible S/PDIF transmitter. A signal generator for controlling haptics devices is included; vibe actuators can connect directly to the PDM output interface. All inputs, outputs, and system interfaces can function concurrently.

A SLIMbus interface supports multichannel audio paths and host control register access. Three further digital audio interfaces are provided, each supporting a wide range of standard audio sample rates and serial interface formats. Automatic sample-rate detection enables seamless wideband/narrowband voice-call handover. Two FLLs are integrated, providing support for a wide range of system-clock frequencies.

The CS42L92 is configured using the SLIMbus, SPI[™], or I²C interfaces. The device is powered from 1.8- and 1.2-V supplies. The power, clocking, and output driver architectures are designed to maximize battery life in voice, music, and standby modes. Low-power (10 µA) Sleep Mode is supported, with configurable wake-up events.



Table of Contents

1 Pin Descriptions4	4.4
1.1 WLCSP Pinout	4.5
1.2 Pin Descriptions	4.6
2 Typical Connection Diagram	4.7
3 Characteristics and Specifications	4.8
Table 3-1. Parameter Definitions 10	4.9
Table 3-2. Absolute Maximum Ratings 10	4.10
Table 3-3. Recommended Operating Conditions 11	4.1
Table 3-4. Analog Input Signal Level—IN1xx, IN2xx 11	4.12
Table 3-5. Analog Input Pin Characteristics 11	4.13
Table 3-6. Analog Input Gain—Programmable Gain Amplifiers (PGAs) 11	4.14
Table 3-7. Digital Input Signal Level—DMICDATn 12	4.1
Table 3-8. Output Characteristics 12	4.10
Table 3-9. Input/Output Path Characteristics 12	4.1
Table 3-10. Digital Input/Output 15	4.18
Table 3-11. Miscellaneous Characteristics 16	4.19
Table 3-12. Device Reset Thresholds 17	4.20
Table 3-13. System Clock and Frequency-Locked Loop (FLL) 17	4.2
Table 3-14. Digital Microphone (DMIC) Interface Timing 18	4.2
Table 3-15. Digital Speaker (PDM) Interface Timing 19	4.23
Table 3-16. Digital Audio Interface—Master Mode 20	
Table 3-17. Digital Audio Interface—Slave Mode 21	5 App
Table 3-18. Digital Audio Interface Timing—TDM Mode 21	5.1
Table 3-19. Control Interface Timing—Two-Wire (I2C) Mode 22	5.2
Table 3-20. Control Interface Timing—Four-Wire (SPI) Mode 23 Table 3-21. State of the stat	5.3
Table 3-21. SLIMbus Interface Timing 24	5.4
Table 3-22. JTAG Interface Timing 25	5.5
Table 3-23. Typical Power Consumption 26 Table 2-24. Typical Circuit Literature 26	6 Reg
Table 3-24. Typical Signal Latency 26	7 The
4 Functional Description	8 Pacl
4.1 Overview	9 Orde
4.2 Input Signal Path	10 Rei
4.3 Digital Core	11 Rev

 4.4 DSP Firmware Control 4.5 DSP Peripheral Control 4.6 Digital Audio Interface 4.7 Digital Audio Interface Control 4.8 SLIMbus Interface 4.9 SLIMbus Control Sequences 4.10 SLIMbus Interface Control 4.11 Output Signal Path 4.12 External Accessory Detection 4.13 Low Power Sleep Configuration 4.14 General-Purpose I/O 	. 106 . 123 . 130 . 140 . 142 . 144 . 152 . 169 . 189 . 190
4.15 Interrupts 4.16 Clocking and Sample Rates	. 200
4.17 Control Interface	219
4.18 Control-Write Sequencer	244
4.19 Charge Pumps, Regulators, and Voltage Reference	. 257
4.20 JTAG Interface	. 263
4.21 Short-Circuit Protection	. 263
4.22 Power-On Reset (POR)	. 264
4.23 Hardware Reset, Software Reset, Wake-Up, and	
	. 264
5 Applications	. 20/
5.1 Recommended External Components	. 201
5.3 Output-Signal Drive-Strength Control	274
5.4 Digital Audio Interface Clocking Configurations	276
5.5 PCB Layout Considerations	. 280
6 Register Map	. 281
7 Thermal Characteristics	. 332
8 Package Dimensions	. 333
9 Ordering Information	. 333
10 References	
11 Revision History	. 334



1 Pin Descriptions

1.1 WLCSP Pinout

	A2		A4		A6		A8		A10		A12	
B1	CPVDD1	B3	CP1VOUT2	N B5	HPOUT4L	B7	HPOUT3L	B9	HPOUT2L	B11	HPOUT1L	B13
CP2CA	C2	CPGND	C4	CP1C2B	C6	HPOUT3R	C8	HPOUT2R	C10	HPOUT1R	C12	MICDET4/ HPOUTFB4
D1	CP1C2A	D 3	CP1VOUT2	P D5	HPOUT4R	D7	MICDET5/ HPOUTFB5	D9	HPDET2	D11	MICDET3/ HPOUTFB3	D13
CP2CB	E2	CPVDD2	E4	CP1VOUT1F	· E6	TRST	E8	JACKDET1	E10	HPDET1	E12	GPSW1N
F1	CP1C1A	F3	CP1VOUT1	N F5	MICBIAS2A	F7	GPSW2N	F9	MICDET2/ HPOUTFB2	F11	GPSW1P	F13
CP2VOUT	G2	CP1C1B	G4	MICBIAS1A	G6	TDO	G8	GPSW2P	G10	MICDET1/ HPOUTFB1	G12	JACKDET2
H1	MICBIAS1B	НЗ	MICBIAS10	C H5	MICBIAS2B	H7	CIFMODE	Н9	JACKDET3	H11	AGND2	H13
MICVDD	J2	MICBIAS1	D J4	IN1ALN/ DMICCLK1	J6	GPIO2	J8	GPIO1	J10	IRQ	J12	AVDD2
К1	IN1BLN	КЗ	IN1ALP/ DMICDAT	1 K5	AIF3LRCLK GPIO16	К7	AIF3TXDAT/ GPIO13	К9	SPKCLK/ GPIO3	K11	RESET	K13
IN1BLP	L2	IN1ARN/ DMICCLK	3 L4	AIF1TXDAT/ GPIO5	L6	AIF2LRCLK GPIO12		CIFSS	L10	SPKDAT/ GPIO4	L12	SUBGND
M1	IN1ARP/ DMICDAT3	M3	IN2BR	M5	AIF1RXDAT GPIO7	/ M7	CIFMISO	M9	MCLK3	M11	FLLGND	M13
IN1BR	N2	IN2ARN/ DMICCLK4	4 N4	VREFC	N6	AIF2BCLK GPIO10	/ N8	SLIMCLK	N10	MCLK2	N12	FLLVDD
P1	IN2ARP/ DMICDAT4	P3	IN2ALN/ DMICCLK2		AIF1BCLK/ GPIO6	P7	AIF3BCLK/ GPIO14	P9	SLIMDAT1	P11	DGND	P13
IN2BL	R2	IN2ALP/ DMICDAT	2 R4			AIF2RXDAT GPIO11		CIFSCLK	R10	MCLK1	R12	
(T1) NC	AGND1	T3 AVDD1	SUBGND	T5 DCVDD	AIF1LRCLK GPIO8	AIF2TXDAT GPIO9	AIF3RXDAT/ GPIO15 T/ C		SLIMDAT2 DA	DBVDD	DGND	T13 DGND
			(Analog I		Digital		Power				

Figure 1-1. Top-Down (Through-Package) View—104-Ball WLCSP Package



1.2 Pin Descriptions

Table 1-1 describes each pin on the CS42L92. Note that pins that share a common name should be tied together on the printed circuit board (PCB).

Table 1-1. Pin Descriptions

PU = Pull-up, PD = Pull-down, K = Bus keeper, H = Hysteresis on CMOS input, Z = Hi-Z (High impedance), C = CMOS, OD = Open drain.

Pin Name	Pin #	Power Supply	I/O	Pin Description	Digital Pad Attributes	State at Reset ¹
				Analog I/O		
CP1C1A	E2	_	0	Charge Pump 1 fly-back capacitor 1 pin		
CP1C1B	F3	—	0	Charge Pump 1 fly-back capacitor 1 pin	—	
CP1C2A	C2	—	0	Charge Pump 1 fly-back capacitor 2 pin	—	
CP1C2B	B5	—	0	Charge Pump 1 fly-back capacitor 2 pin	—	
CP1VOUT1N	E4	—	0	Charge Pump 1 negative output 1 decoupling pin	—	Output
CP1VOUT1P	D5	—	0	Charge Pump 1 positive output 1 decoupling pin	—	Output
CP1VOUT2N	A4	_	0	Charge Pump 1 negative output 2 decoupling pin	—	Output
CP1VOUT2P	C4	—	0	Charge Pump 1 positive output 2 decoupling pin	_	Output
CP2CA	B1	—	0	Charge Pump 2 fly-back capacitor pin	—	Output
CP2CB	D1	—	0	Charge Pump 2 fly-back capacitor pin	—	Output
CP2VOUT	F1	_	0	Charge Pump 2 output decoupling pin/supply for LDO2	_	Output
GPSW1N	D13	_	I/O	General-purpose bidirectional switch 1 contact	_	_
GPSW1P	E12	_	I/O	General-purpose bidirectional switch 1 contact	_	_
GPSW2N	E8	_	I/O	General-purpose bidirectional switch 2 contact	_	_
GPSW2P	F9	_	I/O	General-purpose bidirectional switch 2 contact	_	_
HPDET1	D11	_	I/O	Headphone sense 1 input	_	Input
HPDET2	C10	_	I/O	Headphone sense 2 input	_	Input
HPOUT1L	A12	_	0	Left headphone 1 output	_	Output
HPOUT1R	B11	_	0	Right headphone 1 output	_	Output
HPOUT2L	A10	_	0	Left headphone 2 output		Output
HPOUT2R	B9	_	0	Right headphone 2 output		Output
HPOUT3L	A8	_	0	Left headphone 3 output	_	Output
HPOUT3R	B7	_	0	Right headphone 3 output		Output
HPOUT4L	A6	_	0	Left headphone 4 output	_	Output
HPOUT4R	C6	_	0	Right headphone 4 output	_	Output
IN1ALN/ DMICCLK1	H5	MICVDD or MICBIASn [2]	Ι	Left/right-channel negative differential mic/line input/ DMIC Clock Output 1. Also suitable for connection to external accessory interfaces.	—	IN1ALN input
IN1ALP/ DMICDAT1	J4	MICVDD or MICBIASn ^[2]	I	Left-channel single-ended mic/line input/positive differential mic/line input/DMIC Data Input 1. Also suitable for connection to external accessory interfaces.	PD/H	IN1ALP input
IN1ARN/ DMICCLK3	K3	MICVDD or MICBIASn ^[2]	Ι	Right-channel negative differential mic/line input/DMIC Clock Output 3. Also suitable for connection to external accessory interfaces.		IN1ARN input
IN1ARP/ DMICDAT3	L2	MICVDD or MICBIASn ^[2]	I/O	Right-channel single-ended mic/line input/positive differential mic/line input/DMIC Data Input 3. Also suitable for connection to external accessory interfaces.	PD/H	IN1ARP input
IN1BLN	J2	MICVDD	Ι	Negative differential mic/line input. Also suitable for connection to external accessory interfaces.	_	Input
IN1BLP	K1	MICVDD	Ι	Single-ended mic/line input/positive differential mic/line input. Also suitable for connection to external accessory interfaces.	_	Input



Table 1-1. Pin Descriptions (Cont.)

PU = Pull-up, PD = Pull-down, K = Bus keeper, H = Hysteresis on CMOS input, Z = Hi-Z (High impedance), C = CMOS, OD = Open drain.

Dia Nama	D: #	Power	1/0		Digital Pad	04-44 D
Pin Name	Pin #	Supply	I/O	Pin Description	Attributes	State at Reset ¹
IN1BR	M1	MICVDD	Ι	Right-channel single-ended mic/line input/positive differential mic/line input.	_	Input
IN2ALN/ DMICCLK2	N4	MICVDD or MICBIASn [2]	Ι	Left-channel negative differential mic/line input/DMIC Clock Output 2. Also suitable for connection to external accessory interfaces.	—	IN2ALN input
IN2ALP/ DMICDAT2	P3	MICVDD or MICBIASn ^[2]	I/O	Left-channel single-ended mic/line input/positive differential mic/line input/DMIC Data Input 2. Also suitable for connection to external accessory interfaces.	PD/H	IN2ALP input
IN2ARN/ DMICCLK4	M3	MICVDD or MICBIASn ^[2]	I	Right-channel negative differential mic/line input/DMIC Clock Output 4. Also suitable for connection to external accessory interfaces.	_	IN2ARN input
IN2ARP/ DMICDAT4	N2	MICVDD or MICBIASn [2]	I/O	Right-channel single-ended mic/line input/positive differential mic/line input/DMIC Data Input 4. Also suitable for connection to external accessory interfaces.	PD/H	IN2ARP input
IN2BL	P1	MICVDD	Ι	Left-channel single-ended mic/line input	_	Input
IN2BR	L4	MICVDD	Ι	Right-channel single-ended mic/line input. Also suitable for connection to external accessory interfaces.	—	Input
JACKDET1	D9	AVDD	Ι	Jack detect input 1	—	Input
JACKDET2	F13	AVDD	I	Jack detect input 2	_	Input
JACKDET3	G10	AVDD	I	Jack detect input 3	_	Input
MICBIAS1A	F5	MICVDD	0	Microphone bias 1A	_	Output
MICBIAS1B	G2	MICVDD	0	Microphone bias 1B	_	Output
MICBIAS1C	G4	MICVDD	0	Microphone bias 1C	_	Output
MICBIAS1D	H3	MICVDD	0	Microphone bias 1D	_	Output
MICBIAS2A	E6	MICVDD	0	Microphone bias 2A	_	Output
MICBIAS2B	G6	MICVDD	0	Microphone bias 2B	—	Output
MICDET1/ HPOUTFB1	F11	—	I/O	Mic/accessory sense input 1/HPOUT ground feedback pin 1		Input
MICDET2/ HPOUTFB2	E10	—	I/O	Mic/accessory sense input 2/HPOUT ground feedback pin 2		Input
MICDET3/ HPOUTFB3	C12	—	I/O	Mic/accessory sense input 3/HPOUT ground feedback pin 3		Input
MICDET4/ HPOUTFB4	B13		I/O	Mic/accessory sense input 4/HPOUT ground feedback pin 4		Input
MICDET5/ HPOUTFB5	C8		I/O	Mic/accessory sense input 5/HPOUT ground feedback pin 5		Input
MICVDD	H1	_	0	LDO2 output decoupling pin (generated internally by CS42L92). (Can also be used as reference/supply for external microphones.)	_	Output
VREFC	M5	_	0	Band-gap reference external capacitor connection	_	Output
				Digital I/O		
AIF1BCLK/ GPIO6	N6	DBVDD	I/O	Audio interface 1 bit clock/GPIO6	PU/PD/K/H/ Z/C/OD	GPIO6 input with bus-keeper
AIF1LRCLK/ GPIO8	R6	DBVDD	I/O	Audio interface 1 left/right clock/GPIO8	PU/PD/K/H/ Z/C/OD	GPIO8 input with bus-keeper
AIF1RXDAT/ GPIO7	L6	DBVDD	I/O	Audio interface 1 RX digital audio data/GPIO7	PU/PD/K/H/ C/OD	GPIO7 input with bus-keeper
AIF1TXDAT/ GPIO5	K5	DBVDD	I/O	Audio interface 1 TX digital audio data/GPIO5	PU/PD/K/H/ Z/C/OD	GPIO5 input with bus-keeper



Table 1-1. Pin Descriptions (Cont.)

PU = Pull-up, PD = Pull-down, K = Bus keeper, H = Hysteresis on CMOS input, Z = Hi-Z (High impedance), C = CMOS, OD = Open drain.

Pin Name	Pin #	Power Supply	I/O	Pin Description	Digital Pad Attributes	State at Reset ¹
AIF2BCLK/ GPIO10	M7	DBVDD	I/O	Audio interface 2 bit clock/GPIO10	PU/PD/K/H/ Z/C/OD	GPIO10 input with bus-keeper
AIF2LRCLK/ GPIO12	K7	DBVDD	I/O	Audio interface 2 left/right clock/GPIO12	PU/PD/K/H/ Z/C/OD	GPIO12 input with bus-keeper
AIF2RXDAT/ GPIO11	P7	DBVDD	I/O	Audio interface 2 RX digital audio data/GPIO11	PU/PD/K/H/ C/OD	GPIO11 input with bus-keeper
AIF2TXDAT/ GPIO9	Τ7	DBVDD	I/O	Audio interface 2 TX digital audio data/GPIO9	PU/PD/K/H/ Z/C/OD	GPIO9 input with bus-keeper
AIF3BCLK/ GPIO14	N8	DBVDD	I/O	Audio interface 3 bit clock/GPIO14. If the JTAG interface is configured, this pin provides the TCK input connection.	PU/PD/K/H/ Z/C/OD	GPIO14 input with bus-keeper
AIF3LRCLK/ GPIO16	J6	DBVDD	I/O	Audio interface 3 left/right clock/GPIO16	PU/PD/K/H/ Z/C/OD	GPIO16 input with bus-keeper
AIF3RXDAT/ GPIO15	R8	DBVDD	I/O	Audio interface 3 RX digital audio data/GPIO15. If the JTAG interface is configured, this pin provides the TDI input connection.	PU/PD/K/H/ C/OD	GPIO15 input with bus-keeper
AIF3TXDAT/ GPIO13	J8	DBVDD	I/O	Audio interface 3 TX digital audio data/GPIO13. If the JTAG interface is configured, this pin provides the TMS input connection.	PU/PD/K/H/ Z/C/OD	GPIO13 input with bus-keeper
CIFMISO	L8	DBVDD	0	Control interface (SPI) Master In Slave Out data. The CIFMISO is high impedance if CIF1SS is not asserted.	Z/C	Output
CIFMOSI/SDA	Т9	DBVDD	I/O	Control interface (SPI) Master Out Slave In data/ Control interface (I ² C) data input/output.	H/OD	Input
CIFSCLK	P9	DBVDD	I	Control interface clock input	Н	Input
CIFSS	K9	DBVDD	I	Control interface (SPI) slave select (SS)	Н	Input
CIFMODE	G8	DBVDD	I	Control interface mode select	Н	Input
GPIO1	H9	DBVDD	I/O	GPI01	PU/PD/K/H/ C/OD	GPIO1 input with bus-keeper
GPIO2	H7	DBVDD	I/O	GPIO2	PU/PD/K/H/ C/OD	GPIO2 input with bus-keeper
IRQ	H11	DBVDD	0	Interrupt request (IRQ) output (default is active low)	C/OD	Output
MCLK1	P11	DBVDD	I	Master clock 1	Н	Input
MCLK2	M11	DBVDD	Ι	Master clock 2	Н	Input
MCLK3	L10	DBVDD	Ι	Master clock 3	Н	Input
RESET	J12	DBVDD	I	Digital reset input (active low)	PU/PD/K/H	Input with pull-up
SLIMCLK	M9	DBVDD	I/O	SLIMbus clock I/O	H/C	Input
SLIMDAT1	N10	DBVDD	I/O	SLIMbus data I/O	H/C	Input
SLIMDAT2	R10	DBVDD	I/O	SLIMbus data I/O	H/C	Input
SPKCLK/ GPIO3	J10	DBVDD	I/O	Digital speaker (PDM) clock output/GPIO3	PU/PD/K/H/ C/OD	GPIO3 input with bus-keeper
SPKDAT/ GPIO4	K11	DBVDD	I/O	Digital speaker (PDM) data output/GPIO4	PU/PD/K/H/ C/OD	GPIO4 input with bus-keeper
TDO	F7	DBVDD	0	JTAG data output	С	Output
TRST	D7	DBVDD	Ι	JTAG test access port reset (active low)	PD/H	Input with pull-down
				Supply		
AGND1	R2	—	—	Analog ground (return path for AVDD1)	—	—
AGND2	G12		_	Analog ground (return path for AVDD2)	_	
AVDD1	Т3		_	Analog supply		
AVDD2	H13		_	Analog supply		



Table 1-1. Pin Descriptions (Cont.)

PU = Pull-up, PD = Pull-down, K = Bus keeper, H = Hysteresis on CMOS input, Z = Hi-Z (High impedance), C = CMOS, OD = Open drain.

Pin Name	Pin #	Power Supply	I/O	Pin Description	Digital Pad Attributes	State at Reset ¹
CPGND	B3	—	_	Charge pump ground (return path for CPVDD1, CPVDD2)	—	_
CPVDD1	A2	—	—	Supply for Charge Pump 1 and Charge Pump 2	—	_
CPVDD2	D3	—	—	Secondary supply for Charge Pump 1	—	_
DBVDD	T11	_		Digital buffer (I/O) supply	—	_
DCVDD	P13, T5	—	—	Digital core supply	—	_
DGND	N12, P5, R12, T13	_	_	Digital ground (return path for DCVDD and DBVDD)	—	—
FLLGND	L12	_	—	Analog ground (return path for FLLVDD)	—	_
FLLVDD	M13	—	—	Analog FLL supply	—	_
SUBGND	K13, R4	—	_	Substrate ground	—	_
				No Connect		
NC	T1	—		-		_

1. Note that the default conditions described are not valid if modified by the boot sequence or by a wake-up control sequence.

2. The analog input functions on these pins are referenced to the MICVDD power domain. The digital input/output functions are referenced to the MICVDD or MICBIAS*n* power domain, as selected by the applicable INx_DMIC_SUP field.



2 Typical Connection Diagram

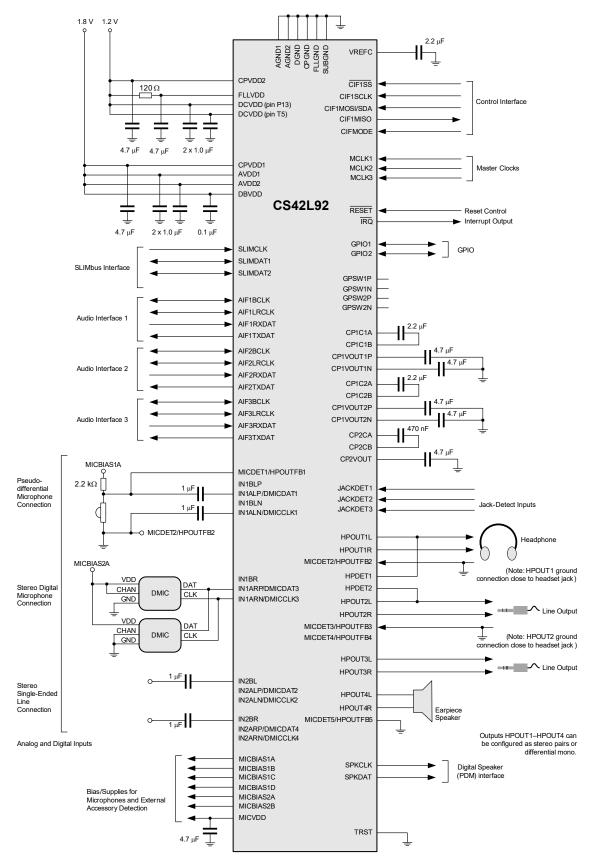


Figure 2-1. Typical Connection Diagram



Characteristics and Specifications 3

Table 3-1 defines parameters as they are characterized in this section.

Table 3-1. Parameter Definitions

Parameter	Definition
Channel separation	Left-to-right and right-to-left channel separation is the difference in level between the active channel (driven to maximum full scale output) and the measured signal level in the idle channel at the test signal frequency. The active channel is configured and supplied with an appropriate input signal to drive a full scale output, with signal measured at the output of the associated idle channel.
Common-mode rejection ratio (CMRR)	The ratio of a specified input signal (applied to both sides of a differential input), relative to the output signal that results from it.
Dynamic range (DR)	A measure of the difference between the maximum full scale output signal and the sum of all harmonic distortion products plus noise, with a low-level input signal applied. Typically, an input signal level 60 dB below full scale is used.
Power-supply rejection ratio (PSRR)	The ratio of a specified power supply variation relative to the output signal that results from it. PSRR is measured under quiescent signal path conditions.
Signal-to-noise ratio (SNR)	A measure of the difference in level between the maximum full scale output signal and the output with no input signal applied.
Total harmonic distortion (THD)	The ratio of the RMS sum of the harmonic distortion products in the specified bandwidth ¹ relative to the RMS amplitude of the fundamental (i.e., test frequency) output.
Total harmonic distortion plus noise (THD+N)	The ratio of the RMS sum of the harmonic distortion products plus noise in the specified bandwidth ¹ relative to the RMS amplitude of the fundamental (i.e., test frequency) output.

1.All performance measurements are specified with a 20-kHz, low-pass brick-wall filter and, where noted, an A-weighted filter. The low-pass filter removes out-of-band noise.

Table 3-2. Absolute Maximum Ratings

Absolute maximum ratings are stress ratings only. Permanent damage to the device may be caused by continuously operating at or beyond these limits. Device functional operating limits and guaranteed performance specifications are given under electrical characteristics at the test conditions specified.

Parameter	Symbol	Minimum	Maximum
Supply voltages	DCVDD ^[1] , FLLVDD ^[1]	–0.3 V	1.6 V
	AVDD ^[2] , CPVDD1, CPVDD2	–0.3 V	2.5 V
	DBVDD, MICVDD	–0.3 V	5.0 V
Voltage range digital inputs DBVDD domain DMICDAT1-DMICDAT2		SUBGND – 0.3 V SUBGND – 0.3 V	DBVDD + 0.3 V MICVDD + 0.3 V
Voltage range analog inputs	IN1ARx, IN2Axx, IN2Bx IN1ALx, IN1BLx, IN1BR MICDET <i>n</i> ³ HPOUTFB <i>n</i> ³ JACKDET1, HPDET1, HPDET2 JACKDET2 ^[4] , JACKDET3 ^[4] GPSW <i>n</i> P, GPSW <i>n</i> N	SUBGND – 0.3 V SUBGND – 0.9 V SUBGND – 0.3 V SUBGND – 0.3 V CP1VOUT2N – 0.3 V SUBGND – 0.3 V SUBGND – 0.3 V	MICVDD + 0.3 V MICVDD + 0.3 V MICVDD + 0.3 V SUBGND + 0.3 V AVDD + 0.3 V MICVDD + 0.3 V MICVDD + 0.3 V
Ground	AGND ⁶ , DGND, CPGND, FLLGND	SUBGND – 0.3V	SUBGND + 0.3V
Operating temperature range	T _A	-40°C	+85°C
Operating junction temperature	TJ	-40°C	+125°C
Storage temperature after soldering		-65°C	+150°C



ESD-sensitive device. The CS42L92 is manufactured on a CMOS process. It is therefore generically susceptible to damage from excessive static voltages. Proper ESD precautions must be taken during handling and storage of this device. This device is gualified to current JEDEC ESD standards.

1. The DCVDD and FLLVDD pins should be tied to a common supply rail. The associated power domain is referred to as DCVDD.

2. The AVDD1 and AVDD2 pins should be tied together. The associated power domain is referred to as AVDD.

3. The MICDET n and HPOUTFB n functions share common pins. The absolute maximum rating varies according to the applicable function of each pin. The HPOUTFBn ratings are applicable if any of the HPn GND SEL bits select the respective pin for HPOUT ground feedback.

4.If AVDD > MICVDD (e.g., if LDO2 is disabled), the maximum JACKDET2/JACKDET3 voltage is AVDD + 0.3 V.

5.CP1VOUT2N is an internal supply, generated by the CS42L92 charge pump (CP1). Its voltage can vary between CPGND and -CPVDD1.

6. The AGND1 and AGND2 pins should be tied together. The associated ground domain is referred to as AGND.



Table 3-3. Recommended Operating Conditions

Parameter		Symbol	Minimum	Typical	Maximum	Units
Digital supply range ¹	Core and FLL	DCVDD [2], FLLVDD [3]	1.14	1.2	1.26	V
Digital supply range	I/O	DBVDD	1.71		3.6 [4]	V
Charge pump supply range	CPVDD1	CPVDD1	1.71	1.8	1.89	V
	CPVDD2	CPVDD2	1.14	1.2	1.26	V
Analog supply range ^{5,6}		AVDD	1.71	1.8	1.89	V
Mic bias supply ⁷		MICVDD	0.9	2.5	3.78	V
Ground ⁸		DGND, AGND, CPGND, FLLGND, SUBGND	—	0	_	V
Power supply rise time 9,10		DCVDD	10	_	2000	μs
		All other supplies	10	—		μs
Operating temperature range		T _A	-40	_	85	°C

Note: There are no power sequencing requirements; the supplies may be enabled and disabled in any order.

1. The DCVDD and FLLVDD pins should be tied to a common supply rail. The associated power domain is referred to as DCVDD.

2. Sleep mode is supported for when DCVDD is below the limits noted, provided that AVDD and DBVDD are present.

3.It is recommended to connect a 120-Ω resistor in series with the FLLVDD pin connection. Note that the minimum voltage limit applies at the supply end of the 120-Ω resistor in this case.

4. If the SLIMbus interface is enabled, the maximum DBVDD voltage is 1.98 V.

5. The AVDD1 and AVDD2 pins should be tied together. The associated power domain is referred to as AVDD.

6. The AGND1 and AGND2 pins should be tied together. The associated ground domain is referred to as AGND.

7.An internal charge pump and LDO (powered by CPVDD1) provide the mic bias supply; the MICVDD pin must not be connected to an external supply. 8.The impedance between DGND, AGND, CPGND, FLLGND, and SUBGND must not exceed 0.1 Ω.

9.If the DCVDD rise time exceeds 2 ms, RESET must be asserted during the rise and held asserted until after DCVDD is within the recommended operating limits.

10. The specified minimum power supply rise times assume a minimum decoupling capacitance of 100 nF per pin. However, Cirrus Logic strongly advises that the recommended decoupling capacitors are present on the PCB and that appropriate layout guidelines are observed. The specified minimum power supply rise times also assume a maximum PCB inductance of 10 nH between decoupling capacitor and pin.

Table 3-4. Analog Input Signal Level—IN1xx, IN2xx

Test conditions (unless specified otherwise): AVDD = 1.8V; with the exception of the condition noted, the following electrical characteristics are valid across the full range of recommended operating conditions.

Parame	Minimum	Typical	Maximum	Units	
Full-scale input signal level (0 dBFS output)	Single-ended PGA input, 0 dB PGA gain	—	0.5	—	V _{RMS}
		—	-6	—	dBV
	Differential PGA input, 0 dB PGA gain	—	1	_	V _{RMS}
		—	0	—	dBV

Notes:

The full-scale input signal level is also the maximum analog input level, before clipping occurs.

- The maximum input signal level is reduced by 6 dB if mid-power operation is selected (IN*n*_OSR = 100); the maximum signal level corresponds to –6 dBFS at the respective ADC outputs in this case.
- The full-scale input signal level changes in proportion with AVDD. For differential input, it is calculated as AVDD / 1.8.
- A 1.0V_{RMS} differential signal equates to 0.5V_{RMS}/–6dBV per input.
- · A sinusoidal input signal is assumed.

Table 3-5. Analog Input Pin Characteristics

Test conditions (unless specified otherwise): $T_A = +25^{\circ}C$; with the exception of the condition noted, the following electrical characteristics are valid across the full range of recommended operating conditions.

Parameter			Typical	Maximum	Units
Input resistance	Single-ended PGA input, All PGA gain settings	9	11	_	kΩ
	Differential PGA input, All PGA gain settings	17	22	—	kΩ
Input capacitance		—	_	5	pF

Table 3-6. Analog Input Gain—Programmable Gain Amplifiers (PGAs)

The following electrical characteristics are valid across the full range of recommended operating conditions.

Parameter		imum	Typical	Maximum	Units
Minimum programmable gain	-	_	0	—	dB
Maximum programmable gain	-	_	31	—	dB
Programmable gain step size Guaranteed mon	otonic		1		dB



Table 3-7. Digital Input Signal Level—DMICDATn

The following electrical characteristics are valid across the full range of recommended operating conditions.

Full-scale input level 1 0 dBFS digital core input, 0 dB gain 6 dBFS	Parameter	Minimum	Typical	Maximum	Units
			-6	—	dBFS

1. The digital input signal level is measured in dBFS, where 0 dBFS is a signal level equal to the full-scale range (FSR) of the PDM input. The FSR is defined as the amplitude of a 1-kHz sine wave whose positive and negative peaks are represented by the maximum and minimum digital codes respectively—this is the largest 1-kHz sine wave that can fit in the digital output range without clipping.

Table 3-8. Output Characteristics

The following electrical characteristics are valid across the full range of recommended operating conditions.

	Parameter		Minimum	Typical	Maximum	Units
Line/headphone/earpiece	Load resistance	Normal operation, Single-Ended Mode	6	_	—	Ω
output driver (HPOUT <i>n</i> L,	N	lormal operation, Differential (BTL) Mode	15	—	—	Ω
HPOUT <i>n</i> R)	De	vice survival with load applied indefinitely	0	—	—	Ω
	Load capacitance	Single-Ended Mode	—	_	500	pF
		Differential (BTL) Mode	—	—	200	pF
Digital speaker output (SPKDAT)	Full-scale output level ¹	0 dBFS digital core output, 0 dB gain	—	6	—	dBFS

1. The digital output signal level is measured in dBFS, where 0 dBFS is a signal level equal to the full-scale range (FSR) of the PDM output. The FSR is defined as the amplitude of a 1-kHz sine wave whose positive and negative peaks are represented by the maximum and minimum digital codes respectively—this is the largest 1-kHz sine wave that can fit in the digital output range without clipping.

Table 3-9. Input/Output Path Characteristics

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = 3.1 V (powered from internal LDO); $T_A = +25^{\circ}C$; 1 kHz sinusoid signal; Fs = 48 kHz; PGA gain = 0 dB, 24-bit audio data.

	Parameter		Min	Тур	Мах	Units
	DC offset at Load	Single-ended mode		50	—	μV
driver (HPOUT <i>n</i> L, HPOUT <i>n</i> R)		Differential (BTL) mode	_	75	—	μV
Analog input paths (IN1xx,	SNR (A-weighted), defined in Table 3-1	20 Hz to 20 kHz, 48 kHz sample rate		99		dB
IN2xx) to ADC (Differential Input		20 Hz to 8 kHz, 16 kHz sample rate		104	—	dB
Mode)	THD, defined in Table 3-1	–1 dBV input	_	-87	—	dB
	THD+N, defined in Table 3-1	–1 dBV input	Ι	-88	-79	dB
	Channel separation (L/R), defined in Table 3-	1 100 Hz to 10 kHz	—	109	—	dB
	Input-referred noise floor	A-weighted, PGA gain = +20 dB	_	2.6	—	μV_{RMS}
	CMRR, defined in Table 3-1	PGA gain = +30 dB	—	83	—	dB
		PGA gain = 0 dB	—	72	—	dB
	PSRR (DBVDD, CPVDD1, AVDD), defined	100 mV (peak-peak) 217 Hz		99		dB
	in Table 3-1	100 mV (peak-peak) 10 kHz	_	84		dB
	PSRR (DCVDD, FLLVDD, CPVDD2),	100 mV (peak-peak) 217 Hz		100	_	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		82	—	dB
Analog input paths (IN1xx,	SNR (A-weighted), defined in Table 3-1	20 Hz to 20 kHz, 48 kHz sample rate	87	98	—	dB
IN2xx) to ADC (Single-Ended		20 Hz to 8 kHz, 16 kHz sample rate		108	—	dB
Input Mode)	THD, defined in Table 3-1	–7dB V input		-84	—	dB
	THD+N, defined in Table 3-1	–7dB V input	—	-83	-78	dB
	Channel separation (L/R), defined in Table 3-	1 100 Hz to 10 kHz	Ι	107		dB
	Input-referred noise floor	A-weighted, PGA gain = +20 dB	—	4	—	μV_{RMS}
	PSRR (DBVDD, CPVDD1, AVDD), defined	100 mV (peak-peak) 217 Hz	_	76		dB
	in Table 3-1	100 mV (peak-peak) 10 kHz	—	52	—	dB
	PSRR (DCVDD, FLLVDD, CPVDD2),	100 mV (peak-peak) 217 Hz		96	_	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz	_	87	—	dB



Table 3-9. Input/Output Path Characteristics (Cont.)

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = 3.1 V (powered from internal LDO); $T_A = +25^{\circ}C$; 1 kHz sinusoid signal; Fs = 48 kHz; PGA gain = 0 dB, 24-bit audio data.

	Parameter		Min	Тур	Max	Units
Analog input paths (IN1xx,	SNR, defined in Table 3-1	A-weighted		88	—	dB
IN2xx) to ADC (Differential	THD, defined in Table 3-1	–7 dBV input		-81		dB
Input, Mid Power Mode)	THD+N, defined in Table 3-1	–7 dBV input		-80		dB
	Channel separation (L/R), defined in Table 3-1	100 Hz to 10 kHz	—	96		dB
	Input-referred noise floor	A-weighted, PGA gain = +20 dB	—	4.79		μV_{RMS}
	CMRR, defined in Table 3-1	PGA gain = +30 dB PGA gain = 0 dB		73 81		dB dB
	PSRR (DBVDD, CPVDD1, AVDD), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz		93 75		dB dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz	—	97 81	_	dB dB
DAC to line output (HPOUT1 x , HPOUT2 x ; Load = 10 k Ω ,	Full-scale output signal level	0 dBFS input		1 0	_	V _{RMS} dBV
50 pF)	SNR, defined in Table 3-1	A-weighted, output signal = $1 V_{RMS}$	_	125	_	dB
	Dynamic range, defined in Table 3-1	A-weighted, –60 dBFS input		117	_	dB
	THD+N, defined in Table 3-1	0 dBFS input		-100	-90	dB
	Channel separation (L/R), defined in Table 3-1	· · · · · · · · · · · · · · · · · · ·		95		dB
	Output noise floor	A-weighted		0.7	_	μV _{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		105		dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz	—	81	_	dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz	_	108 81	_	dB dB
DAC to headphone output	Maximum output power	0.1% THD+N		33	_	mW
(HPOUT1 <i>x</i> , HPOUT2 <i>x</i> ;	SNR, defined in Table 3-1	A-weighted, output signal = $1 V_{RMS}$	_	125	_	dB
$R_L = 32 \Omega$)	Dynamic range, defined in Table 3-1	A-weighted, -60 dBFS input		117	_	dB
	THD+N, defined in Table 3-1	$P_0 = 20 \text{ mW}$		-100	-90	dB
	THD+N, defined in Table 3-1	$P_0 = 20 \text{ mW}$ $P_0 = 2 \text{ mW}$		-96	-90	dB
	Channel separation (L/R), defined in Table 3-1	-		-90 105	_	dB
	Output noise floor	A-weighted		0.6	_	
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz				μV _{RMS} dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz	—	126 103	_	dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz	—	128 105	_	dB dB
DAC to headphone output	Maximum output power	0.1% THD+N	—	46	—	mW
(HPOUT1 <i>x</i> , HPOUT2 <i>x</i> ; $R_{L} = 16 \Omega$)	SNR, defined in Table 3-1	A-weighted, output signal = 1 V_{RMS}	—	125	_	dB
10 22)	Dynamic range, defined in Table 3-1	A-weighted, –60 dBFS input		117		dB
	THD+N, defined in Table 3-1	P _O = 20 mW		-97	-90	dB
	THD+N, defined in Table 3-1	$P_0 = 2 \text{ mW}$		-95	—	dB
	Channel separation (L/R), defined in Table 3-1			97		dB
	Output noise floor	A-weighted		0.6		μV_{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		127		dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		107	_	dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz		127 108	_	dB dB
DAC to headphone output	Maximum output power	0.1% THD+N		100	_	mW
(HPOUT1L+HPOUT1R,	SNR, defined in Table 3-1	A-weighted, output signal = $2 V_{RMS}$		103		dB
HPOUT2L+HPOUT2R; Stereo	Dynamic range, defined in Table 3-1	A-weighted, -60 dBFS input		115	_	dB
differential output, $R_L = 32 \Omega$	THD+N, defined in Table 3-1	$P_{\Omega} = 75 \text{ mW}$		-100	_	dB
BTL)	THD+N, defined in Table 3-1	$P_0 = 75 \text{ mW}$ $P_0 = 5 \text{ mW}$		-100 -97		dB
	Output noise floor	A-weighted		0.36		
	PSRR (DBVDD, CPVDD1, AVDD),					μV _{RMS}
	defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz	—	120 90	_	dB dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz		120 90	_	dB dB



Table 3-9. Input/Output Path Characteristics (Cont.)

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = 3.1 V (powered from internal LDO); $T_A = +25^{\circ}C$; 1 kHz sinusoid signal; Fs = 48 kHz; PGA gain = 0 dB, 24-bit audio data.

	Parameter		Min		Мах	
DAC to line output (HPOUT3 x , HPOUT4 x ; Load = 10 k Ω ,	Full-scale output signal level	0 dBFS input	_	1 0	_	V _{RMS} dBV
50 pF)	SNR, defined in Table 3-1	A-weighted, output signal = $1 V_{RMS}$	115	125	—	dB
	Dynamic range, defined in Table 3-1	A-weighted, -60 dBFS input		115	—	dB
	THD, defined in Table 3-1	0 dBFS input		-92	—	dB
	THD+N, defined in Table 3-1	0 dBFS input		-91	-85	dB
	Channel separation (L/R), defined in Table 3-1	100 Hz to 10 kHz		90	_	dB
	Output noise floor	A-weighted		0.63	_	μV_{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		110	—	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		78	—	dB
	PSRR (DCVDD, FLLVDD, CPVDD2),	100 mV (peak-peak) 217 Hz		99		dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		69	—	dB
DAC to headphone output	Maximum output power	0.1% THD+N	_	33	-	mW
(HPOUT3x, HPOUT4x;	SNR, defined in Table 3-1	A-weighted, output signal = $1 V_{RMS}$		124	-	dB
R _L = 32 Ω)	Dynamic range, defined in Table 3-1	A-weighted, -60 dBFS input	107	115		dB
	THD, defined in Table 3-1	P _O = 20 mW		-93	—	dB
	THD+N, defined in Table 3-1	P _O = 20 mW	—	-90	-85	dB
	THD, defined in Table 3-1	P _O = 2 mW		-94	—	dB
	THD+N, defined in Table 3-1	P _O = 2 mW	—	-92	—	dB
	Channel separation (L/R), defined in Table 3-1	100 Hz to 10 kHz	—	102	—	dB
	Output noise floor	A-weighted	_	0.63	—	μV_{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		127		dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		100	—	dB
	PSRR (DCVDD, FLLVDD, CPVDD2),	100 mV (peak-peak) 217 Hz		120	-	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		95	—	dB
DAC to headphone output (HPOUT3 <i>x</i> , HPOUT4 <i>x</i> ;	Maximum output power	0.1% THD+N	—	46	—	mW
$(\text{R}_{\text{L}} = 16 \Omega)$	SNR, defined in Table 3-1	A-weighted, output signal = $1 V_{RMS}$		124	—	dB
11 - 10 32)	Dynamic range, defined in Table 3-1	A-weighted, -60 dBFS input		115	—	dB
	THD, defined in Table 3-1	P _O = 20 mW		-94	—	dB
	THD+N, defined in Table 3-1	P _O = 20 mW		-89	-80	dB
	THD, defined in Table 3-1	$P_0 = 2 \text{ mW}$		-91	—	dB
	THD+N, defined in Table 3-1	$P_0 = 2 \text{ mW}$		-89	—	dB
	Channel separation (L/R), defined in Table 3-1			100	—	dB
	Output noise floor	A-weighted		0.63	—	μV_{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		128	—	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		104	_	dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz		125 97	—	dB dB
DAC to correigoo output		0.1% THD+N		97 115		mW
DAC to earpiece output (HPOUT3L+HPOUT3R,	Maximum output power SNR, defined in Table 3-1		—		—	
HPOUT4L+HPOUT4R, Mono		A-weighted, output signal = $2 V_{RMS}$		129	_	dB
Mode, $R_L = 32 \Omega BTL)$	Dynamic range, defined in Table 3-1 THD, defined in Table 3-1	A-weighted, –60 dBFS input		120	_	dB
		P _O = 75 mW		-99	—	dB
	THD+N, defined in Table 3-1	P _O = 75 mW		-97		dB
	THD, defined in Table 3-1	$P_0 = 5 \text{ mW}$		-98		dB
	THD+N, defined in Table 3-1	$P_0 = 5 \text{ mW}$		-94		dB
	Output noise floor	A-weighted		0.36	—	μV _{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		127	—	dB dB
		100 mV (peak-peak) 10 kHz 100 mV (peak-peak) 217 Hz		106		dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1	100 mV (peak-peak) 217 Hz 100 mV (peak-peak) 10 kHz		125 101		dB dB
			_	101		чD



Table 3-9. Input/Output Path Characteristics (Cont.)

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = 3.1 V (powered from internal LDO); $T_A = +25^{\circ}C$; 1 kHz sinusoid signal; Fs = 48 kHz; PGA gain = 0 dB, 24-bit audio data.

	Parameter		Min	Тур	Мах	Units
DAC to earpiece output	Maximum output power	0.1% THD+N	_	138	—	mW
	SNR, defined in Table 3-1	A-weighted, output signal = 2 V _{RMS}	_	128	—	dB
HPOUT4L+HPOUT4R, Mono Mode, R_L = 16 Ω BTL)	Dynamic range, defined in Table 3-1	A-weighted, –60 dBFS input	110	120	-	dB
	THD, defined in Table 3-1	P _O = 75 mW	_	-97	-	dB
	THD+N, defined in Table 3-1	P _O = 75 mW		-95	-	dB
	THD, defined in Table 3-1	P _O = 5 mW		-96	-	dB
	THD+N, defined in Table 3-1	P _O = 5 mW	_	-94	-	dB
	Output noise floor	A-weighted		0.4	-	μV_{RMS}
	PSRR (DBVDD, CPVDD1, AVDD),	100 mV (peak-peak) 217 Hz		125	_	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz		106	—	dB
	PSRR (DCVDD, FLLVDD, CPVDD2),	100 mV (peak-peak) 217 Hz		125	—	dB
	defined in Table 3-1	100 mV (peak-peak) 10 kHz	—	101	—	dB

Table 3-10. Digital Input/Output

The following electrical characteristics are valid across the full range of recommended operating conditions.

	Parameter		Minimum	Typical	Maximum	Units
Digital I/O (except	Input HIGH level	V _{DBVDD} = 1.71–1.98 V	$0.75 \times DBVDD$	_	—	V
DMICDAT <i>n</i> and		$V_{DBVDD} = 2.5 V \pm 10\%$		—	—	V
DMICCLKn) 1,2		$V_{DBVDD} = 3.3 V \pm 10\%$	$0.7 \times \text{DBVDD}$	_	—	V
	Input LOW level	V _{DBVDD} = 1.71–1.98 V		_	$0.3 \times DBVDD$	V
		$V_{DBVDD} = 2.5 V \pm 10\%$		—	$0.25 \times DBVDD$	V
		V_{DBVDD} = 3.3 V ±10%		_	$0.2 \times DBVDD$	V
	Output HIGH level	V _{DBVDD} = 1.71–1.98 V		—	—	V
	(I _{OH} = 1 mA)	V _{DBVDD} = 2.5 V ±10%		—	—	V
		V _{DBVDD} = 3.3 V ±10%	$0.7 \times \text{DBVDD}$	_	—	V
	Output LOW level	V _{DBVDDn} = 1.71–1.98 V	_	_	$0.25 \times DBVDD$	V
	(I _{OL} = 1mA)	$V_{DBVDDn} = 2.5 V \pm 10\%$	—	—	$0.3 \times \text{DBVDD}$	V
		$V_{\text{DBVDD}n} = 3.3 \text{ V} \pm 10\%$	_	_	$0.15 \times DBVDD$	V
	Input capacitance		_	_	5	pF
	Input leakage		-10		10	μA
	Pull-up/pull-down resistance (where ap	plicable)	35	_	55	kΩ
DMIC I/O	DMICDAT <i>n</i> input HIGH level		$0.65 \times V_{SUP}$	_	—	V
(DMICDAT <i>n</i> and	DMICDAT <i>n</i> input LOW level		—	—	$0.35 \times V_{SUP}$	V
DMICCLK <i>n</i>) ^{2,3}	DMICCLKn output HIGH level	I _{OH} = 1 mA	$0.8 \times V_{SUP}$	_	—	V
	DMICCLKn output LOW level	I _{OL} = -1 mA	_		$0.2 \times V_{SUP}$	V
	Input capacitance		—	25		pF
	Input leakage		-1	_	1	μA
GPIO <i>n</i>	Clock output frequency GPIO pir	as OPCLK or FLL output	_	—	50	MHz

1. Digital I/O is referenced to DBVDD.

2. Note that digital input pins should not be left unconnected or floating.

3. DMICDAT *n* and DMICCLK*n* are referenced to a selectable supply, V_{SUP}, according to the IN*n*_DMIC_SUP fields.



Table 3-11. Miscellaneous Characteristics

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = 3.1 V (powered from internal LDO); T_A = +25°C; 1 kHz sinusoid signal; Fs = 48 kHz; PGA gain = 0 dB, 24-bit audio data.

	Parameter	Min	Тур	Max	Units
Microphone bias	Minimum bias voltage ²		1.5	—	V
(MICBIASnx) ¹	Maximum bias voltage	—	2.8	-	V
	Bias voltage output step size		0.1		V
	Bias voltage accuracy	-5%	_	+5%	V
	Bias current ³ Regulator Mode (MICB <i>n</i> _BYPASS = 0), V _{MICVDD} – V _{MICBIAS} >200 mV	—	—	2.4	mA mA
	Bypass Mode (MICBn_BYPASS = 1) Output noise density Regulator Mode (MICBn BYPASS = 0), MICBn LVL = 0x4,	_		5.0	
	Output noise density Regulator Mode (MICBn_BYPASS = 0), MICBn_LVL = 0x4, Load current = 1 mA, Measured at 1 kHz	—	50		nV/√Hz
	Integrated noise voltage Regulator Mode (MICBn BYPASS = 0), MICBn LVL = 0x4,	_	4	_	μV _{RMS}
	Load current = 1 mA, 100 Hz to 7 kHz, A-weighted	_	4		μvRMS
	PSRR (DBVDD, CPVDD1, AVDD), defined in Table 3-1 100 mV (peak-peak) 217 Hz		105	_	dB
	100 mV (peak-peak) 10 kHz	_	95	_	dB
	PSRR (DCVDD, FLLVDD, CPVDD2), defined in Table 3-1 100 mV (peak-peak) 217 Hz		99	_	dB
	100 mV (peak-peak) 10 kHz	_	92	_	dB
	Load capacitance ³ Regulator Mode (MICBn BYPASS = 0), MICBn EXT CAP = 0	_	_	50	pF
	Regulator Mode (MICBn_BYPASS = 0), MICBn_EXT_CAP = 1	0.1	1.0	10	μF
	Output discharge resistance MICBnx_ENA = 0, MICBnx_DISCH = 1		2	—	kΩ
General-purpose		_	40	—	Ω
switch 4	Switch open	—	100	—	MΩ
External	Headphone detection load impedance range: HPD_IMPEDANCE_RANGE = 00	4	_	30	Ω
Accessory Detect	Detection via HPDET1 (HPD_SENSE	8	—	100	Ω
	SEL = 0100) or HPDET2 (HPD_SENSE		—	1000	Ω
	SEL = 0101) HPD_IMPEDANCE_RANGE = 11		_	10000	Ω
	Headphone detection load impedance range: Detection via MICDET <i>n</i> or JACKDET <i>n</i> pins	400		6000	Ω
	Headphone detection accuracy: HPD_IMPEDANCE_RANGE = 01 or 10	-5		+5	%
	(HPD_DACVAL, HPDET <i>n</i> pin) HPD_IMPEDANCE_RANGE = 00 or 11		-	+10	%
	Headphone detection accuracy (HPD_LVL, MICDET <i>n</i> or JACKDET <i>n</i> pin)	-20	_	+20	%
	Microphone impedance detection range: for MICDn_LVL[0] = 1	0	—	70	Ω
	$(MICDn_ADC_MODE = 0, 2.2 k\Omega \pm 2\% MICBIAS resistor. 5) $ for MICDn_LVL[1] = 1		—	180	Ω
	for MICD <i>n</i> _LVL[2] = 1 for MICD <i>n</i> _LVL[3] = 1	210 360	_	290 680	Ω Ω
	for MICDn_LVL[8] = 1		_	30000	Ω
	Jack-detection input threshold voltage Detection on JACKDET1, Jack insertion		0.9	_	V
	(JACKDET <i>n</i>) Detection on JACKDET1, Jack removal	_	1.65	_	V
	Detection on JACKDET2/3, Jack insertion	—	0.27	—	V
	Detection on JACKDET2/3, Jack removal	—	0.9	—	V
MICVDD Charge	Output voltage	0.9	2.7	3.3	V
Pump and	Programmable output voltage step size LDO2_VSEL = 0x00–0x14 (0.9–1.4V)		25	—	mV
Regulator (CP2	LDO2_VSEL = 0x14 to 0x27 (1.4 V-3.3 V)		100	—	mV
and LDO2)	Maximum output current		8	—	mA
	Start-up time 4.7 µF on MICVDD	_	1.0	2.5	ms
Frequency-Lock	Output frequency	45		50	MHz
ed Loop (FLL1,	Lock Time F _{REF} = 32 kHz, F _{FLL} = 49.152 MHz	_	5	—	ms
FLL2)	F _{REF} = 12 MHz, F _{FLL} = 49.152 MHz		1		ms
RESET pin input	RESET input pulse width 6	1	—	—	μs
				1	

1.No capacitor on MICBIASnx. In Regulator Mode, it is required that V_{MICVDD} - V_{MICBIAS} > 200 mV.

2.Regulator Mode (MICB $n_BYPASS = 0$), Load current ≤ 1.0 mA.

3. Bias current and load capacitance specifications are per MICBIAS generator (MICBIAS1 or MICBIAS2). 4. The GPSW*n*N pin voltage must not exceed GPSW*n*P + 0.3 V. See Table 3-2 for voltage limits applicable to the GPSW*n*P and GPSW*n*N pins.

5. These characteristics assume no other component is connected to MICDETn.

6. To trigger a hardware reset, the RESET input must be asserted for longer than this duration.



Table 3-12. Device Reset Thresholds

The following electrical characteristics are valid across the full range of recommended operating conditions.

Parame	ter	Symbol	Minimum	Typical	Maximum	Units
AVDD reset threshold	V _{AVDD} rising V _{AVDD} falling	V _{AVDD}	 1.06	_	1.66 1.44	V V
DCVDD reset threshold	V _{DCVDD} rising V _{DCVDD} falling	V _{DCVDD}	 0.49	_	1.04 0.70	V V
DBVDD Reset threshold	V _{DBVDD} rising V _{DBVDD} falling	V _{DBVDD}	 1.06	_	1.66 1.44	V V

Note: The reset thresholds are derived from simulations only, across all operational and process corners. Device performance is not assured outside the voltage ranges defined in Table 3-3.

Table 3-13. System Clock and Frequency-Locked Loop (FLL)

The following timing information is valid across the full range of recommended operating conditions.

	Pa	rameter	Minimum	Typical	Maximum	Units
Master clock	MCLK cycle time	MCLK as input to FLL, FLLn_REFCLK_DIV = 00	74		_	ns
timing (MCLK1,	-	MCLK as input to FLL, FLLn_REFCLK_DIV = 01	37	—	—	ns
MCLK2, MCLK3) ¹		MCLK as input to FLL, FLLn_REFCLK_DIV = 10	18	—	—	ns
		MCLK as input to FLL, FLLn_REFCLK_DIV = 11	12.5	—	—	ns
		MCLK as direct SYSCLK or ASYNCCLK source	40		—	ns
	MCLK duty cycle	MCLK as input to FLL	80:20	_	20:80	%
		MCLK as direct SYSCLK or ASYNCCLK source	60:40		40:60	%
Frequency-locked	FLL input frequency	FLLn_REFCLK_DIV = 00	0.032	_	13	MHz
loop (FLL1, FLL2)		FLLn_REFCLK_DIV = 01	0.064	—	26	MHz
		FLLn_REFCLK_DIV = 11	0.128	—	52	MHz
		FLLn_REFCLK_DIV = 11	0.256		80	MHz
Internal clocking	SYSCLK frequency	SYSCLK_FREQ = 000, SYSCLK_FRAC = 0	-1%	6.144	+1%	MHz
		SYSCLK_FREQ = 000, SYSCLK_FRAC = 1	-1%	5.6448	+1%	MHz
		SYSCLK_FREQ = 001, SYSCLK_FRAC = 0	-1%	12.288	+1%	MHz
		SYSCLK_FREQ = 001, SYSCLK_FRAC = 1	-1%	11.2896	+1%	MHz
		SYSCLK_FREQ = 010, SYSCLK_FRAC = 0	-1%	24.576	+1%	MHz
		SYSCLK_FREQ = 010, SYSCLK_FRAC = 1	-1%	22.5792	+1%	MHz
		SYSCLK_FREQ = 011, SYSCLK_FRAC = 0	-1%	49.152	+1%	MHz
		SYSCLK_FREQ = 011, SYSCLK_FRAC = 1	-1%	45.1584	+1%	MHz
		SYSCLK_FREQ = 100, SYSCLK_FRAC = 0	-1%	98.304	+1%	MHz
		SYSCLK_FREQ = 100, SYSCLK_FRAC = 1	-1%	90.3168	+1%	MHz
	ASYNCCLK frequency	ASYNC_CLK_FREQ = 000	-1%	6.144	+1%	MHz
			-1%	5.6448	+1%	MHz
		ASYNC_CLK_FREQ = 001	-1%	12.288	+1%	MHz
			-1%	11.2896	+1%	MHz
		ASYNC_CLK_FREQ = 010	-1%	24.576	+1%	MHz
			-1%	22.5792	+1%	MHz
		ASYNC_CLK_FREQ = 011	-1%	49.152	+1%	MHz
			-1%	45.1584	+1%	MHz
		ASYNC_CLK_FREQ = 100	-1%	98.304	+1%	MHz
			-1%	90.3168	+1%	MHz
	DSPCLK frequency		5	—	150	MHz

1.If MCLK1, MCLK2, or MCLK3 is selected as a source for SYSCLK or ASYNCCLK (either directly or via one of the FLLs), the frequency must be within 1% of the applicable SYSCLK_FREQ or ASYNC_CLK_FREQ setting.



Table 3-14. Digital Microphone (DMIC) Interface Timing

The following timing information is valid across the full range of recommended operating conditions.

Parameter ¹	Symbol	Minimum	Typical	Maximum	Units
DMICCLKn cycle time	t _{CY}	160	163	1432	ns
DMICCLK <i>n</i> duty cycle	—	45	_	55	%
DMICCLK <i>n</i> rise/fall time (25-pF load, 1.8-V supply)	t _r , t _f	5	_	30	ns
DMICDAT <i>n</i> (left) setup time to falling DMICCLK edge	t _{LSU}	15	—	—	ns
DMICDAT <i>n</i> (left) hold time from falling DMICCLK edge	t _{LH}	0	_	—	ns
DMICDAT <i>n</i> (right) setup time to rising DMICCLK edge	t _{RSU}	15	_	—	ns
DMICDAT <i>n</i> (right) hold time from rising DMICCLK edge	t _{RH}	0	_	—	ns

Note: The voltage reference for the DMIC interfaces is selectable, using the IN*n*_DMIC_SUP fields—each interface may be referenced to MICVDD, MICBIAS1, or MICBIAS2.

1.DMIC interface timing

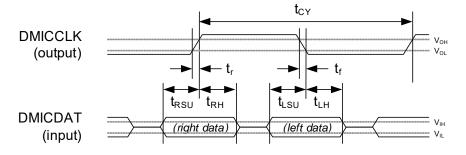


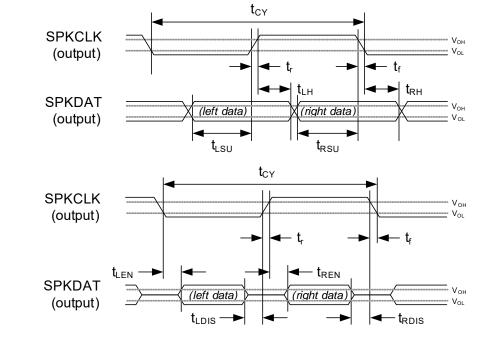


Table 3-15. Digital Speaker (PDM) Interface Timing

The following timing information is valid across the full range of recommended operating conditions.

	Parameter	Symbol	Minimum	Typical	Maximum	Units
Mode A ¹	SPKCLK cycle time	t _{CY}	160	163	358	ns
	SPKCLK duty cycle	—	45	—	55	%
	SPKCLK rise/fall time (25-pF load)	t _r , t _f	2	—	8	ns
	SPKDAT set-up time to SPKCLK rising edge (left channel)	t _{LSU}	30		—	ns
	SPKDAT hold time from SPKCLK rising edge (left channel)	t _{LH}	30	—	—	ns
	SPKDAT set-up time to SPKCLK falling edge (right channel)	t _{RSU}	30	—	—	ns
	SPKDAT hold time from SPKCLK falling edge (right channel)	t _{RH}	30		—	ns
Mode B ²	SPKCLK cycle time	t _{CY}	160	163	358	ns
	SPKCLK duty cycle	—	45	—	55	%
	SPKCLK rise/fall time (25-pF load)	t _r , t _f	2		8	ns
	SPKDAT enable from SPKCLK rising edge (right channel)	t _{REN}	—	—	15	ns
	SPKDAT disable to SPKCLK falling edge (right channel)	t _{RDIS}	—	—	5	ns
	SPKDAT enable from SPKCLK falling edge (left channel)	t _{LEN}	—		15	ns
	SPKDAT disable to SPKCLK rising edge (left channel)	t _{LDIS}	—	—	5	ns

1. Digital speaker (PDM) interface timing—Mode A



2. Digital speaker (PDM) interface timing-Mode B



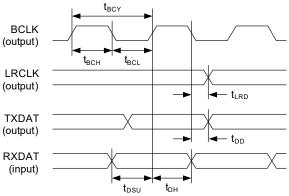
Table 3-16. Digital Audio Interface—Master Mode

Test conditions (unless specified otherwise): C_{LOAD} = 25 pF (output pins); BCLK slew (10% to 90%) = 3.7–5.6 ns; with the exception of the conditions noted, the following electrical characteristics are valid across the full range of recommended operating conditions.

	Parameter ¹			Typical	Maximum	Units
Master Mode	AIF <i>n</i> BCLK cycle time	t _{BCY}	40		—	ns
	AIF <i>n</i> BCLK pulse width high	t _{BCH}	18	—	—	ns
	AIF <i>n</i> BCLK pulse width low	t _{BCL}	18	—	—	ns
	t _{LRD}	0		8.3	ns	
	AIF <i>n</i> TXDAT propagation delay from BCLK falling edge	t _{DD}	0	—	5	ns
	AIF nRXDAT setup time to BCLK rising edge	t _{DSU}	11	—	—	ns
	AIF nRXDAT hold time from BCLK rising edge	t _{DH}	0		—	ns
Master Mode,	AIF nLRCLK setup time to BCLK rising edge	t _{LRSU}	14		—	ns
Slave LRCLK	AIF nLRCLK hold time from BCLK rising edge	t _{LRH}	0		—	ns

Notes: The descriptions above assume noninverted polarity of AIFnBCLK.

1. Digital audio interface timing—Master Mode. Note that BCLK and LRCLK outputs can be inverted if required; the figure shows the default, noninverted polarity.



2. The timing of the AIF*n*LRCLK signal is selectable. If the LRCLK advance option is enabled, the LRCLK transition is timed relative to the preceding BCLK edge. Under the required condition that BCLK is inverted in this case, the LRCLK transition is still timed relative to the falling BCLK edge.



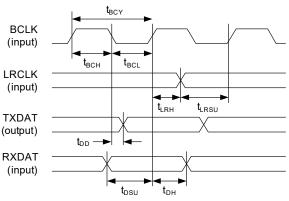
Table 3-17. Digital Audio Interface—Slave Mode

The following timing information is valid across the full range of recommended operating conditions, unless otherwise noted.

	Parameter ^{1,2}	Symbol	Min	Тур	Мах	Units
AIF <i>n</i> BCLK cycle time		t _{BCY}	40	—	—	ns
AIF <i>n</i> BCLK pulse width high	BCLK as direct SYSCLK or ASYNCCLK source	t _{BCH}	16	_	_	ns
	All other conditions	t _{BCH}	14	—	—	ns
AIFnBCLK pulse width low	BCLK as direct SYSCLK or ASYNCCLK source	t _{BCL}	16	—	—	ns
	All other conditions	t _{BCL}	14	—	—	ns
$C_{LOAD} = 15 pF$ (output pins),	AIF <i>n</i> LRCLK set-up time to BCLK rising edge	t _{LRSU}	7			ns
BCLK slew (10%–90%) = 3 ns	AIF nLRCLK hold time from BCLK rising edge	t _{LRH}	0	_	-	ns
	AIF nTXDAT propagation delay from BCLK falling edge	t _{DD}	0		12.2	ns
	AIF nRXDAT set-up time to BCLK rising edge	t _{DSU}	2	_		ns
	AIF nRXDAT hold time from BCLK rising edge	t _{DH}	0	_		ns
	Master LRCLK, AIF nLRCLK propagation delay from BCLK falling edge	t _{LRD}	_		14.8	ns
C _{LOAD} = 25 pF (output pins),	AIF nLRCLK set-up time to BCLK rising edge	t _{LRSU}	7	_		ns
BCLK slew (10%–90%) = 6 ns	AIF nLRCLK hold time from BCLK rising edge	t _{LRH}	0			ns
	AIF nTXDAT propagation delay from BCLK falling edge	t _{DD}	0		14.2	ns
	AIF nRXDAT set-up time to BCLK rising edge	t _{DSU}	2	_		ns
	AIF nRXDAT hold time from BCLK rising edge	t _{DH}	0	_		ns
	Master LRCLK, AIF nLRCLK propagation delay from BCLK falling edge	t _{LRD}	—	—	15.9	ns

Note: The descriptions above assume noninverted polarity of AIF nBCLK.

1. Digital audio interface timing—Slave Mode. Note that BCLK and LRCLK inputs can be inverted if required; the figure shows the default, noninverted polarity.



2.If AIF nBCLK or AIF nLRCLK is selected as a source for SYSCLK or ASYNCCLK (either directly or via one of the FLLs), the frequency must be within 1% of the applicable SYSCLK_FREQ or ASYNC_CLK_FREQ setting.

Table 3-18. Digital Audio Interface Timing—TDM Mode

The following timing information is valid across the full range of recommended operating conditions, unless otherwise noted.

Par	Min	Тур	Max	Units	
Master Mode—C _{LOAD} (AIF <i>n</i> TXDAT) = 15 to	AIF <i>n</i> TXDAT enable time from BCLK falling edge	0	_	—	ns
25 pF. BCLK slew (10%–90%) = 3.7ns to 5.6 ns.	AIF <i>n</i> TXDAT disable time from BCLK falling edge	_	_	6	ns
EGILD	AIF <i>n</i> TXDAT enable time from BCLK falling edge	2	_	—	ns
BCLK slew (10%–90%) = 3 ns	AIF <i>n</i> TXDAT disable time from BCLK falling edge	_	_	12.2	ns
	AIF <i>n</i> TXDAT enable time from BCLK falling edge	2	_	—	ns
BCLK slew (10%–90%) = 6 ns	AIF <i>n</i> TXDAT disable time from BCLK falling edge		_	14.2	ns

Note: If TDM operation is used on the AIF*n*TXDAT pins, it is important that two devices do not attempt to drive the AIF*n*TXDAT pins simultaneously. To support this requirement, the AIF*n*TXDAT pins can be configured to be tristated when not outputting data.

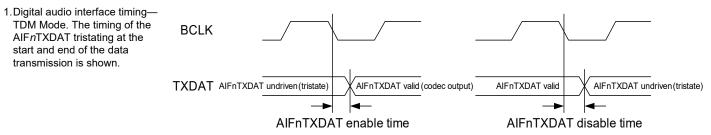




 Table 3-19. Control Interface Timing—Two-Wire (I²C) Mode

 The following timing information is valid across the full range of recommended operating conditions.

	Parameter ¹	Symbol	Min	Тур	Max	Units
SCLK frequency		_	—	—	3400	kHz
SCLK pulse-width low		t ₁	160	—	—	ns
SCLK pulse-width high		t ₂	100			ns
Hold time (start condition)		t ₃	160	—		ns
Setup time (start condition)		t ₄	160	—	—	ns
SDA, SCLK rise time (10%–90%)	SCLK frequency > 1.7 MHz SCLK frequency > 1 MHz	t ₆ t ₆	_	_	80 160	ns ns
	SCLK frequency ≤ 1 MHz	t ₆		_	2000	ns
SDA, SCLK fall time (90%–10%)	SCLK frequency > 1.7 MHz SCLK frequency > 1 MHz SCLK frequency ≤ 1 MHz	t ₇ t ₇ t ₇			60 160 200	ns ns ns
Setup time (stop condition)	· · ·	t ₈	160			ns
SDA setup time (data input)		t ₅	40	—	_	ns
SDA hold time (data input)		t ₉	0	—	—	ns
SDA valid time (data/ACK output)	SCLK slew (90%–10%) = 20ns, C _{LOAD} (SDA) = 15 pF SCLK slew (90%–10%) = 60ns, C _{LOAD} (SDA) = 100 pF SCLK slew (90%–10%) = 160ns, C _{LOAD} (SDA) = 400 pF SCLK slew (90%–10%) = 200ns, C _{LOAD} (SDA) = 550 pF	t ₁₀ t ₁₀ t ₁₀ t ₁₀			40 130 190 220	ns ns ns ns
Pulse width of spikes that are supp		t _{ps}	0	—	25	ns

1. Control interface

timing—I²C Mode

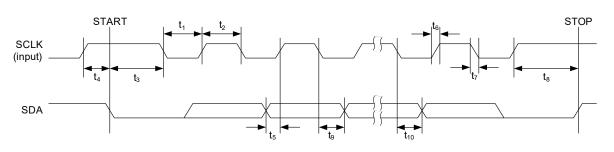


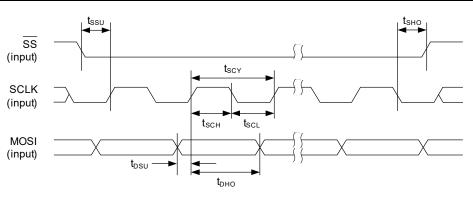


Table 3-20. Control Interface Timing—Four-Wire (SPI) Mode

The following timing information is valid across the full range of recommended operating conditions.

Parameter ^{1, 2}			Min	Тур	Max	Units
SS falling edge to SCLK rising edge		t _{SSU}	2.6	_	_	ns
SCLK falling edge to SS rising edge		t _{SHO}	0	_	_	ns
SCLK pulse cycle time	SYSCLK disabled (SYSCLK_ENA = 0)		38.4	—		ns
	SYSCLK_ENA = 1, SYSCLK_FREQ = 000	t _{SCY}	76.8	—	—	ns
	SYSCLK_ENA = 1, SYSCLK_FREQ > 000	t _{SCY}	38.4		—	ns
SCLK pulse-width low		t _{SCL}	15.3	_		ns
SCLK pulse-width high		t _{SCH}	15.3			ns
MOSI to SCLK set-up time		t _{DSU}	1.5	_	_	ns
MOSI to SCLK hold time		t _{DHO}	1.7	—		ns
SCLK falling edge to MISO transition	SCLK slew (90%–10%) = 5 ns, C _{LOAD} (MISO) = 25 pF	t _{DL}	0	—	12.6	ns

1.Control interface timing—SPI Mode (write cycle)



2.Control interface timing—SPI Mode (read cycle)

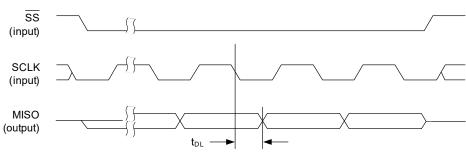




Table 3-21. SLIMbus Interface Timing

The following timing information is valid across the full range of recommended operating conditions.

		Parameter ¹	Symbol	Minimum	Тур	Maximum	Units
SLIMCLK	SLIMCLK cycle ti	SLIMCLK cycle time				—	ns
input	SLIMCLK pulse width high			12	—	—	ns
	SLIMCLK pulse v	vidth low	T _{CLKH} T _{CLKL}	12	—	—	ns
SLIMCLK	SLIMCLK cycle ti	me		40	—	_	ns
output	SLIMCLK pulse v	vidth high	T _{CLKH}	12	—	—	ns
	SLIMCLK pulse v	vidth low	T _{CLKL}	12	—	—	ns
	SLIMCLK slew	C _{LOAD} = 15 pF, SLIMCLK_DRV_STR = 0		0.09 x V _{DBVDD}	—	0.22 x V _{DBVDD}	V/ns
	rate (20%-80%)	C _{LOAD} = 70 pF, SLIMCLK_DRV_STR = 0	SR _{CLK}	0.02 x V _{DBVDD}	—	0.05 x V _{DBVDD}	V/ns
		C _{LOAD} = 70 pF, SLIMCLK_DRV_STR = 1	SR _{CLK}	0.04 x V _{DBVDD}	—	0.11 x V _{DBVDD}	V/ns
SLIMDAT	SLIMDAT setup ti	ime to SLIMCLK falling edge	T _{SETUP}	3.5	—	—	ns
input	SLIMDAT hold tin	ne from SLIMCLK falling edge	Т _Н	2	—	—	ns
SLIMDAT	SLIMDAT time	SLIMDAT_DRV_STR = 0, DBVDD = 1.71 V	T _{DV}	_	4.7	8.1	ns
output	for data output	C _{LOAD} = 15 pF, SLIMDAT_DRV_STR = 1, DBVDD = 1.71 V	T _{DV}	—	4.3	7.3	ns
	valid (relative to	C _{LOAD} = 30 pF, SLIMDAT_DRV_STR = 0, DBVDD = 1.71 V	T _{DV}	—	6.8	11.8	ns
	SLIMCLK rising	C _{LOAD} = 30 pF, SLIMDAT_DRV_STR = 1, DBVDD = 1.71 V	T _{DV}	—	5.8	10.0	ns
	edge)	C _{LOAD} = 50 pF, SLIMDAT_DRV_STR = 0, DBVDD = 1.71 V	T _{DV}	—	9.6	16.6	ns
		C _{LOAD} = 50 pF, SLIMDAT_DRV_STR = 1, DBVDD = 1.71 V	T _{DV}	—	7.9	13.7	ns
		C _{LOAD} = 70 pF, SLIMDAT_DRV_STR = 0, DBVDD = 1.71 V	T _{DV}	—	12.4	21.5	ns
		C _{LOAD} = 70 pF, SLIMDAT_DRV_STR = 1, DBVDD = 1.71 V	T _{DV}	—	10.0	17.4	ns
	SLIMDAT slew	C _{LOAD} = 15 pF, SLIMDAT_DRV_STR = 0	SR _{DATA}	—	—	0.64 x V _{DBVDD}	V/ns
	rate (20%–80%)	$C_{LOAD} = 30 \text{ pF}, \text{SLIMDAT}_DRV_STR = 0$	SR _{DATA}	—	—	0.35 x V _{DBVDD}	V/ns
		C _{LOAD} = 30pF, SLIMDAT_DRV_STR = 1	SR _{DATA}	—	—	0.46 x V _{DBVDD}	V/ns
		C _{LOAD} = 70pF, SLIMDAT_DRV_STR = 0	SR _{DATA}	—	—	0.16 x V _{DBVDD}	V/ns
		C _{LOAD} = 70pF, SLIMCLK_DRV_STR = 1	SR _{DATA}	—	—	0.21 x V _{DBVDD}	V/ns
Other	Driver disable tim		T _{DD}	—	—	6	ns
parameters	Bus holder output	t impedance $0.1 \text{ x V}_{\text{DBVDD}} < \text{V} < 0.9 \text{ x V}_{\text{DBVDD}}$	R _{DATAS}	18	—	50	kΩ

Notes:

• The signal timing information describes the timing requirements of the SLIMbus interface as a whole, not just the CS42L92 device.

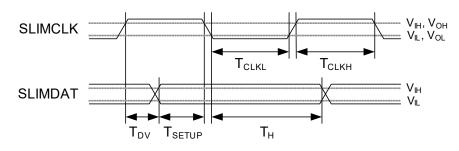
 T_{DV} is the propagation delay from the rising SLIMCLK edge (at CS42L92 input) to the SLIMDAT output being achieved at the input to all devices across the bus.

• T_{SETUP} is the set-up time for SLIMDAT input (at CS42L92), relative to the falling SLIMCLK edge (at CS42L92).

• T_H is the hold time for SLIMDAT input (at CS42L92) relative to the falling SLIMCLK edge (at CS42L92).

• For more details of the interface timing, refer to the MIPI Alliance Specification for Serial Low-Power Inter-Chip Media Bus (SLIMbus)

1.SLIMbus interface timing.



 V_{IL} , V_{IH} are the 35%/65% levels of the respective inputs.

 V_{OL} , V_{OH} are the 20%/80% levels of the respective outputs.

The SLIMDAT output delay (T_{DV}) is with respect to the input pads of all receiving devices



Table 3-22. JTAG Interface Timing

Test conditions (unless specified otherwise): $C_{LOAD} = 25 \text{ pF}$ (output pins); TCK slew (20%–80%) = 5 ns; with the exception of the conditions noted, the following electrical characteristics are valid across the full range of recommended operating conditions.

Parameter ¹	Symbol	Minimum	Typical	Maximum	Units
TCK cycle time	T _{CCY}	50	_		ns
TCK pulse width high	Т _{ССН}	20	_		ns
TCK pulse width low	T _{CCL}	20	_	—	ns
TMS setup time to TCK rising edge	T _{MSU}	1	_		ns
TMS hold time from TCK rising edge	T _{MH}	2	_		ns
TDI setup time to TCK rising edge	T _{DSU}	1	_	—	ns
TDI hold time from TCK rising edge	T _{DH}	2	_		ns
TDO propagation delay from TCK falling edge	T _{DD}	0	_	17	ns
TRST setup time to TCK rising edge	T _{RSU}	3	_	—	ns
TRST hold time from TCK rising edge	T _{RH}	3	_		ns
TRST pulse-width low	_	20		—	ns

1.JTAG Interface timing

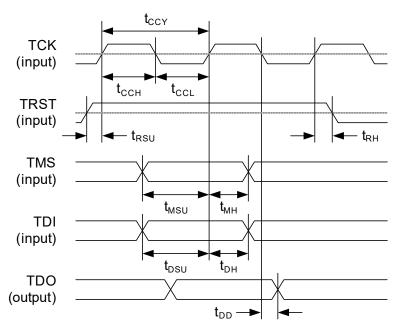




Table 3-23. Typical Power Consumption

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = Off (CP2 and LDO2 disabled); $T_A = +25^{\circ}C$; Fs = 48 kHz; 24-bit audio data, I²S Slave Mode; SYSCLK = 24.576 MHz (direct MCLK1 input).

	Operating Configuration		Typical I _{1.2V} (mA)	Typical I _{1.8V} (mA)	P _{TOT} (mW)
Headphone playback	AIF1 to DAC to HPOUT1 (stereo), 32-Ω load.	Quiescent 1-kHz sine wave, P _O = 10 mW		0.8 1.9	3.6 48.78
Earpiece playback	AIF1 to DAC to HPOUT1 (mono), 32-Ω load (BTL).	Quiescent 1-kHz sine wave, P _O = 30 mW		0.85 1.7	2.97 77.1
Stereo line record	Analog line to ADC to AIF1, MICVDD = 1.8V (CP2 and LDO2 bypass enabled).	1-kHz sine wave, -1 dBFS output	1.1	2.2	5.28
Sleep Mode	Accessory detect enabled (JD1_EN	IA = 1)	0.000	0.010	0.018

Table 3-24. Typical Signal Latency

Test conditions (unless specified otherwise): DBVDD = CPVDD1 = AVDD = 1.8 V, DCVDD = FLLVDD = CPVDD2 = 1.2 V; MICVDD = Off (CP2 and LDO2 disabled); $T_A = +25^{\circ}C$; Fs = 48 kHz; 24-bit audio data, I²S Slave Mode; SYSCLK = 24.576 MHz (direct MCLK1 input).

Operating Configura	tion	Latency (µs)
AIF to DAC path—digital input (AIF <i>n</i>) to analog output (HPOUT <i>n</i>)	192 kHz input, 192 kHz output, Synchronous	237
	96 kHz input, 96 kHz output, Synchronous	269
	48 kHz input, 48 kHz output, Synchronous	358
	44.1 kHz input, 44.1 kHz output, Synchronous	374
	16 kHz input, 16 kHz output, Synchronous	629
	8 kHz input, 8 kHz output, Synchronous	1334
	8 kHz input, 48 kHz output, Isochronous 1	1939
	16 kHz input, 48 kHz output, Isochronous 1	993
	8 kHz input, 44.1 kHz output, Asynchronous ²	1880
	16 kHz input, 44.1 kHz output, Asynchronous ²	1084
ADC to AIF path—analog input (IN <i>n</i>) to digital output (AIF <i>n</i>) ³	192 kHz input, 192 kHz output, Synchronous	50
	96 kHz input, 96 kHz output, Synchronous	96
	48 kHz input, 48 kHz output, Synchronous	193
	44.1 kHz input, 44.1 kHz output, Synchronous	212
	16 kHz input, 16 kHz output, Synchronous	558
	8 kHz input, 8 kHz output, Synchronous	1170
	8 kHz input, 48 kHz output, Isochronous ¹	1696
	16 kHz input, 48 kHz output, Isochronous ¹	861
	44.1 kHz input, 8 kHz output, Asynchronous ²	1364
	44.1 kHz input, 16 kHz output, Asynchronous ²	841

1. Signal is routed via the ISRC function in the isochronous cases only.

2. Signal is routed via the ASRC function in the asynchronous cases only.

3. Digital core high-pass filter is included in the signal path.



4 Functional Description

The CS42L92 is a highly integrated, low-power audio hub codec for mobile telephony and portable devices. It provides flexible, high-performance audio interfacing for handheld devices in a small and cost-effective package. The digital audio interfaces and hi-fi DACs support 32-bit playback, offering audiophile-quality playback at sample rates up to 384 kHz. Native audio playback is possible, concurrent with voice and ultrasonic input paths. Multiple analog outputs support a wide variety of the headphone/accessory configurations, including balanced stereo headphone loads.

4.1 Overview

The CS42L92 block diagram is shown in Fig. 4-1.

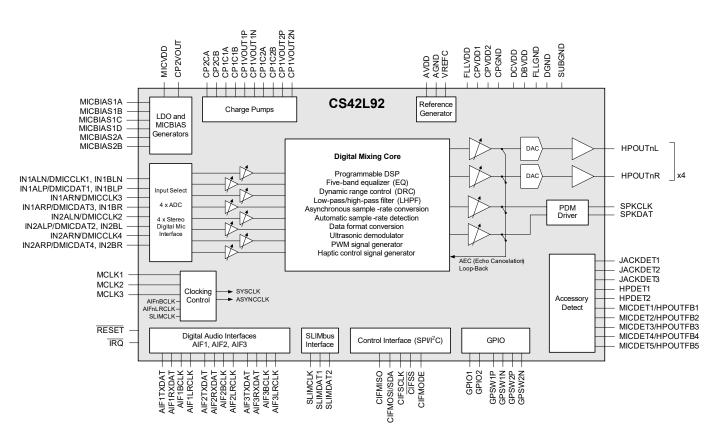


Figure 4-1. CS42L92 Block Diagram

The CS42L92 combines fixed-function signal-processing blocks with a fully-flexible, all-digital audio mixing and routing engine for extensive use-case flexibility. The signal-processing blocks include high-/low-pass filters, EQ, dynamic range control, and sample-rate converters.

The CS42L92 provides multiple digital audio interfaces, including SLIMbus, to provide independent and fully asynchronous connections to different processors (e.g., application processor, baseband processor, and wireless transceiver).

A flexible clocking arrangement supports a wide variety of external clock references, including clocking derived from the digital audio interface. Two frequency-locked loop (FLL) circuits provide additional flexibility for system clocking, including low-power always-on operation. Seamless switching between clock sources is supported, and free-running modes are also available.

Unused circuitry can be disabled under software control to save power; low leakage currents enable extended standby/off time in portable battery-powered applications. The CS42L92 always-on circuitry can be used in conjunction with the applications processor to wake up the device following a headphone jack-detection event.



Versatile GPIO functionality is provided, including support for external accessory/push-button detection inputs. Comprehensive interrupt functions, with status reporting, are also provided. The integrated DSP provides a general-purpose signal processing capability; this is supported by general-purpose timer and event-logger functions.

4.1.1 Hi-Fi Audio Codec

The CS42L92 is a high-performance, low-power audio codec that uses a simple analog architecture. Four ADCs are incorporated, with multiplexers to support up to eight analog inputs. Six DACs are incorporated, with two being switchable between two separate headphone output paths.

The audio codec is controlled directly via register access. The simple analog architecture, combined with the integrated tone generator, enables straightforward device configuration and testing, minimizing debug time and reducing software effort.

The CS42L92 input channels support up to eight analog inputs or up to eight digital inputs, multiplexed into four stereo input signal paths. In differential mode, the analog input path SNR is 104 dB (16-kHz sample rate, i.e., wideband voice mode). The input paths can be configured for low-power operation, ideal for analog or digital microphone input in always-on applications. Ultrasonic signal demodulation functions are provided, supporting a variety of presence-detection applications.

The analog outputs comprise four 33-mW (125 dB SNR) stereo headphone amplifiers with ground-referenced output. The output drivers are designed to support many different system architectures and are compatible with line or headphone loads in single-ended or differential (BTL) configurations. Headphone outputs HPOUT1 and HPOUT2 offer 127 dB SNR and –100 dB THD+N performance with a stereo differential headphone load. Outputs HPOUT3 and HPOUT4 are multiplexed, with the respective DACs and signal paths common to both pairs of output drivers.

Each output path supports independent mixing, equalization, filtering, gain controls. This allows each signal path to be individually tailored for the load characteristics. Selectable hi-fi filters support audiophile playback modes at sample rates up to 384 kHz. All outputs have integrated pop and click suppression features.

The headphone output drivers are ground-referenced, powered from an integrated charge pump, enabling high quality, power efficient headphone playback without any requirement for DC blocking capacitors. Ground loop feedback is incorporated, providing rejection of noise on the ground connections.

The CS42L92 is cost optimized for a wide range of mobile phone applications. External speaker amplifiers can be connected using the stereo PDM outputs; this can ease layout and electromagnetic compatibility by avoiding the need to run high-power speaker outputs over a long distance and across interconnects.

4.1.2 Digital Audio Core

The CS42L92 uses a core architecture based on all-digital signal routing, making digital audio effects available on all signal paths, regardless of whether the source data input is analog or digital. The digital mixing desk allows different audio effects to be applied simultaneously on many independent paths, while supporting a variety of sample rates. Soft mute and unmute control ensures smooth transitions between use cases without interrupting existing audio streams elsewhere.

Highly flexible digital mixing, including mixing between audio interfaces, is possible. The CS42L92 performs multichannel full-duplex asynchronous sample-rate conversion, providing use-case flexibility across a broad range of system architectures. Automatic sample-rate detection is provided, enabling seamless wideband/narrowband voice call handover.

DRC functions are available for optimizing audio signal levels. In playback modes, the DRC can be used to maximize loudness, while limiting the signal level to avoid distortion, clipping, or battery droop, for high-power output drivers such as speaker amplifiers. In record modes, the DRC assists in applications where the signal level is unpredictable.

The five-band parametric EQ functions can be used to compensate for the frequency characteristics of the output transducers. EQ functions can be cascaded to provide additional frequency control. Programmable high-pass and low-pass filters are also available for general filtering applications, such as removal of wind and other low-frequency noise.



4.1.3 Digital Interfaces

Three serial digital audio interfaces (AIFs) each support PCM, TDM, and I²S data formats for compatibility with most industry-standard chipsets. Each AIF supports eight input/output channels. Bidirectional operation at sample rates up to 384 kHz is supported. Data words of up to 32 bits can be routed through AIF1 and AIF3. Data-format conversion (DFC) functions are available to support different interface standards on the input and output signal paths.

Eight digital PDM input channels are available (four stereo interfaces); these are typically used for digital microphones, powered from the integrated MICBIAS power-supply regulators. Two PDM output channels are also available (one stereo interface); these are typically used for external power amplifiers. Embedded mute codes provide a control mechanism for external PDM-input devices.

The auxiliary PDM interface can be used to provide an audio path between an analog microphone connected to the CS42L92 and a digital input to an external audio processor. The auxiliary PDM interface operates in master or slave modes, and is configured on GPIO pins.

The CS42L92 features a SLIMbus interface, compliant with the MIPI[®] SLIMbus specification, providing eight channels of audio input/output. Mixed audio sample rates are supported on the SLIMbus interface. The SLIMbus interface also supports read/write access to the CS42L92 control registers.

An IEC-60958-3–compatible S/PDIF transmitter is incorporated, enabling stereo S/PDIF output on a GPIO pin. Standard S/PDIF sample rates of 32–192 kHz are supported.

Control register access is supported by a configurable SPI/I²C control interface. The interface supports SPI slave operation up to 26 MHz or I²C slave operation up to 3.4 MHz. Full access to the register map is also provided via the SLIMbus port.

4.1.4 Other Features

The CS42L92 incorporates a tone generator that can be used for beep functions through any of the audio signal paths. The tone generator provides two 1-kHz outputs, with configurable phase relationship, offering flexibility to create differential signals or test scenarios.

A white-noise generator is provided that can be routed within the digital core. The noise generator can provide comfort noise in cases where silence (digital mute) is not desirable.

Two pulse-width modulation (PWM) signal generators are incorporated. The duty cycle of each PWM signal can be modulated by an audio source or can be set to a fixed value using a control register setting. The PWM signal generators can be output directly on a GPIO pin.

The CS42L92 supports up to 16 GPIO pins, offering a range of input/output functions for interfacing, for detection of external hardware, and for providing logic outputs to other devices. The CS42L92 provides two dedicated GPIO pins; a further 14 GPIOs are multiplexed with other functions. Comprehensive interrupt functionality is also provided for monitoring internal and external event conditions.

The integrated DSP provides a general-purpose signal processing capability; this is supported by general-purpose timer and event-logger functions.

A signal generator for controlling haptics devices is included, compatible with both eccentric rotating mass (ERM) and linear resonant actuator (LRA) haptics devices. The haptics signal generator is highly configurable and can execute programmable drive event profiles, including reverse drive control. An external vibe actuator can be driven using the PDM digital output path.

A smart accessory interface is included, supporting a wide variety of system configurations. Jack detection, accessory sensing, and impedance measurement is provided, for external headset and push-button detection. Dual headphone connections (e.g., 3.5 mm and USB-C) can be detected simultaneously. Accessory detection can be used as a wake-up trigger from low-power standby. Microphone activity detection with interrupt is also available.



System clocking can be derived from the MCLK1, MCLK2, or MCLK3 input pins. Alternatively, the SLIMbus interface, or the audio interfaces (configured in Slave Mode), can be used to provide a clock reference. The CS42L92 also provides two integrated FLL circuits for clock frequency conversion and stability. The flexible clocking architecture supports low-power always-on operation, with reference frequencies down to 32 kHz. Seamless switching between clock sources is supported; free-running FLL modes are also available.

The CS42L92 is powered from 1.8- and 1.2-V external supplies. Integrated charge-pump and LDO-regulator circuits are used to generate supply rails for internal functions and to support powering or biasing of external microphones. Power consumption is optimized across a wide variety of voice and multimedia use cases.

4.2 Input Signal Path

The CS42L92 provides flexible input channels, supporting up to eight analog inputs or up to eight digital inputs. Selectable combinations of analog (mic or line) and digital inputs are multiplexed into four stereo input signal paths. Input paths IN1 and IN2 support analog and digital inputs; input paths IN3 and IN4 support digital inputs only.

The analog input paths support single-ended and differential configurations, programmable gain control, and are digitized using a high performance sigma-delta ADC. The analog input paths can be configured for low-power operation, ideal for always-on applications. Analog inputs can be configured as input to the auxiliary PDM interface, providing an audio path between an analog microphone connected to the CS42L92 and a digital input to an external audio processor.

The digital input paths interface directly with external digital microphones; a separate microphone interface clock is provided for four separate stereo pairs of digital microphones.

Two microphone bias (MICBIAS) generators provide a low-noise reference for biasing electret condenser microphones (ECMs) or for use as a low-noise supply for MEMS microphones and digital microphones. Switchable outputs from the MICBIAS generators allow six separate reference/supply outputs to be independently controlled.

Digital volume control is available on all inputs (analog and digital), with programmable ramp control for smooth, glitch-free operation. A configurable signal-detect function is available on each input signal path. Ultrasonic signal demodulation functions are provided on the input signal paths, supporting a variety of presence-detection applications.

The input signal paths and control fields are shown in Fig. 4-2.



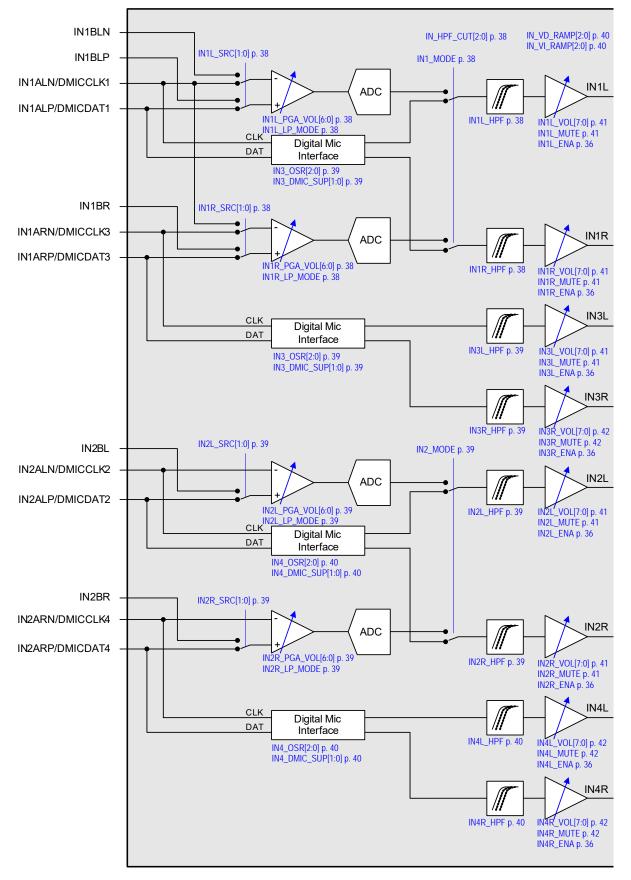


Figure 4-2. Input Signal Paths



4.2.1 Analog Microphone Input

Up to eight analog microphones can be connected to the CS42L92, either in single-ended or differential configuration. The input configuration and pin selection is controlled using IN*nx*_SRC, as described in Section 4.2.6.

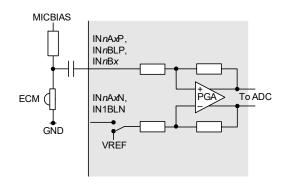
The CS42L92 includes external accessory-detection circuits that can report the presence of a microphone and the status of a hook switch or other push buttons. When using this function, it is recommended to use the IN1AL*x*, IN1BL*x*, or IN1BR analog microphone input paths to ensure best immunity to electrical transients arising from the push buttons.

For single-ended input, the microphone signal is connected to the noninverting input of the PGAs. The inverting inputs of the PGAs are connected to an internal reference in this configuration.

For differential input, the noninverted microphone signal is connected to the noninverting input of the PGAs and the inverted (or noisy ground) signal is connected to the inverting input pins.

The gain of the input PGAs is controlled via register settings, as defined in Section 4.2.6. Note that the input impedance of the analog input paths is fixed across all PGA gain settings.

The ECM analog input configurations are shown in Fig. 4-3 and Fig. 4-4. The integrated MICBIAS generators provide a low noise reference for biasing the ECMs.



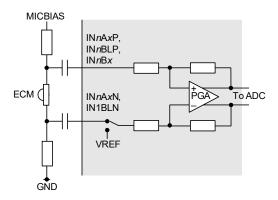


Figure 4-3. Single-Ended ECM Input

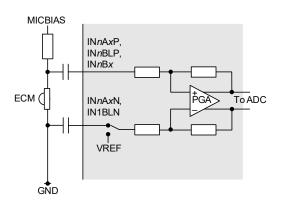
Figure 4-4. Differential ECM Input

Pseudodifferential connection is also possible—this is similar to the configuration shown in Fig. 4-4, but the GND connection is directly to the microphone (and IN*nx*N capacitor), instead of via a resistor. The typical connections for pseudodifferential input are shown in Fig. 4-5.

Note that pseudodifferential input is the recommended configuration if the accessory-detection functions are used on this input path. The IN*nx*_SRC field settings are the same for pseudodifferential connection as for differential.

The IN1ALN pin can be used as the inverting input connection for the IN1L and IN1R paths concurrently. This allows two microphones to be supported, in pseudodifferential configuration, while minimizing the number of pin connections required—see Fig. 4-6.





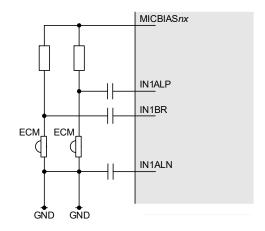


Figure 4-5. Pseudodifferential ECM Input



Analog MEMS microphones can be connected to the CS42L92 in a similar manner to the ECM configurations. Typical configurations are shown in Fig. 4-7 and Fig. 4-8. In this configuration, the integrated MICBIAS generators provide a low-noise power supply for the microphones.

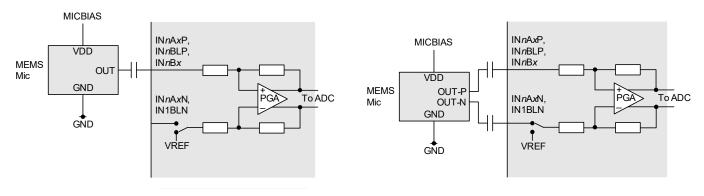
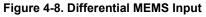


Figure 4-7. Single-Ended MEMS Input



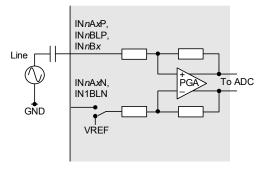
Note: The MICVDD pin can also be used (instead of MICBIAS) as a reference or power supply for external microphones. The MICBIAS outputs are recommended, because they offer better noise performance and independent enable/ disable control.

4.2.2 Analog Line Input

Line inputs can be connected to the CS42L92 in a similar manner to the mic inputs. Single-ended and differential configurations are supported on each analog input path, using the IN*nx*_SRC bits as described in Section 4.2.6.

The analog line input configurations are shown in Fig. 4-9 and Fig. 4-10. Note that the microphone bias (MICBIAS) is not used for line input connections.





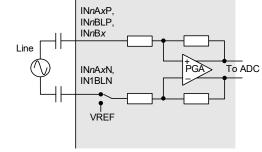


Figure 4-9. Single-Ended Line Input

Figure 4-10. Differential Line Input

4.2.3 DMIC Input

As many as eight digital microphones can be connected to the CS42L92. DMIC operation on input paths IN1–IN4 is selected using IN1_MODE and IN2_MODE, as described in Section 4.2.6.

In DMIC mode, two channels of audio data are multiplexed on the associated DMICDAT*n* pin. Each stereo DMIC interface is clocked using the respective DMICCLK*n* output.

If DMIC input is enabled, the CS42L92 outputs the DMIC clock on the applicable DMICCLK*n* pins. The DMICCLK*n* frequency is controlled by the respective IN*n*_OSR field, as described in Table 4-1 and Table 4-3.

Note that, if the 384- or 768-kHz DMICCLK*n* frequency is selected, the maximum valid sample rate for the respective paths is restricted as described in Table 4-1. If the input sample rates are set globally using IN_RATE (i.e., IN_RATE_MODE = 0), all input paths are affected similarly.

Note that SYSCLK must be present and enabled when using the DMIC inputs; see Section 4.16 for details of SYSCLK and the associated registers.

The DMIC clock frequencies in Table 4-1 assume that the SYSCLK frequency is a multiple of 6.144 MHz (SYSCLK_FRAC = 0). If the SYSCLK frequency is a multiple of 5.6448 MHz (SYSCLK_FRAC = 1), the DMIC clock frequencies are scaled accordingly.

Condition	DMIC Clock Frequency	Valid Sample Rates	Signal Passband
IN <i>n</i> _OSR = 010	384 kHz	Up to 48 kHz	Up to 4 kHz
IN <i>n</i> _OSR = 011	768 kHz	Up to 96 kHz	Up to 8 kHz
IN <i>n</i> _OSR = 100	1.536 MHz	Up to 192 kHz	Up to 20 kHz
IN <i>n</i> _OSR = 101	3.072 MHz	Up to 192 kHz	Up to 20 kHz
IN <i>n</i> _OSR = 110	6.144 MHz	Up to 192 kHz	Up to 96 kHz

Table 4-1. DMICCLK Frequency

The voltage reference for the DMIC interfaces is selectable, using IN*n*_DMIC_SUP; each interface may be referenced to MICVDD, MICBIAS1, or MICBIAS2. The voltage reference for each digital input path should be set equal to the applicable power supply of the respective microphones.

A pair of digital microphones is connected as shown in Fig. 4-11. The microphones must be configured to ensure that the left mic transmits a data bit when DMICCLK is high and the right mic transmits a data bit when DMICCLK is low. The CS42L92 samples the DMIC data at the end of each DMICCLK phase. Each microphone must tristate its data output when the other microphone is transmitting.

Note that the CS42L92 provides integrated pull-down resistors on the DMICDAT*n* pins. This provides a flexible capability for interfacing with other devices.



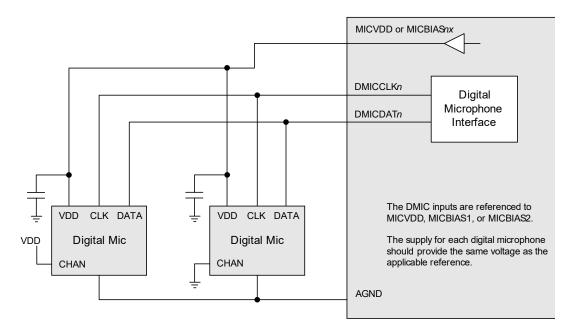


Figure 4-11. DMIC Input

Two DMIC channels are interleaved on DMICDAT*n*. The DMIC interface timing is shown in Fig. 4-12. Each microphone must tristate its data output when the other microphone is transmitting.

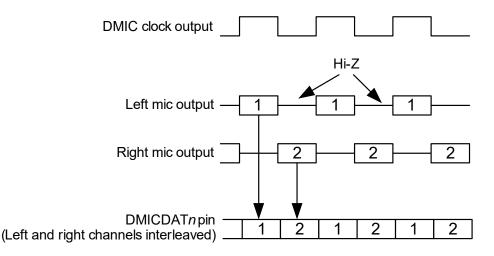


Figure 4-12. DMIC Interface Timing

4.2.4 Input Signal Path Enable

The input signal paths are enabled using the bits described in Table 4-2. The respective bits must be enabled for analog or digital input on the respective input paths.

The input signal paths are muted by default. It is recommended that deselecting the mute should be the final step of the path enable control sequence. Similarly, the mute should be selected as the first step of the path-disable control sequence. The input signal path mute functions are controlled using the bits described in Table 4-4.

The MICVDD power domain must be enabled when using the analog input signal paths. This power domain is provided using an internal charge pump (CP2) and LDO regulator (LDO2). See Section 4.19 for details of these circuits.

The system clock, SYSCLK, must be configured and enabled before any audio path is enabled. The ASYNCCLK and 32-kHz clock may also be required, depending on the path configuration. See Section 4.16 for details of the system clocks.



The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the input signal paths and associated ADCs. If the frequency is too low, an attempt to enable an input signal path fails. Note that active signal paths are not affected under such circumstances.

The status bits in Register R769 indicate the status of each input signal path. If an underclocked error condition occurs, these bits indicate which input signal paths have been enabled.

Register Address	Bit	Label	Default	Description
R768 (0x0300)	7	IN4L_ENA	0	Input Path 4 (left) enable
Input_Enables				0 = Disabled
				1 = Enabled
	6	IN4R ENA	0	Input Path 4 (right) enable
		_		0 = Disabled
				1 = Enabled
	5	IN3L_ENA	0	Input Path 3 (left) enable
		_		0 = Disabled
				1 = Enabled
	4	IN3R_ENA	0	Input Path 3 (right) enable
		_		0 = Disabled
				1 = Enabled
	3	IN2L_ENA	0	Input Path 2 (left) enable
	Ŭ		ů	0 = Disabled
				1 = Enabled
	2	IN2R ENA	0	Input Path 2 (right) enable
	-		Ŭ	0 = Disabled
				1 = Enabled
	1	IN1L_ENA	0	Input Path 1 (left) enable
	I		0	0 = Disabled
				1 = Enabled
	0	IN1R_ENA	0	Input Path 1 (right) enable
	0		0	0 = Disabled
				1 = Enabled
R769 (0x0301)	7	IN4L ENA STS	0	Input Path 4 (left) enable status
	1	IN4L_ENA_515	0	0 = Disabled
Input_Enables_Status				1 = Enabled
			0	
	6	IN4R_ENA_STS	0	Input Path 4 (right) enable status 0 = Disabled
				-
			0	1 = Enabled
	5	IN3L_ENA_STS	0	Input Path 3 (left) enable status
				0 = Disabled
			0	1 = Enabled
	4	IN3R_ENA_STS	0	Input Path 3 (right) enable status
				0 = Disabled
				1 = Enabled
	3	IN2L_ENA_STS	0	Input Path 2 (left) enable status
				0 = Disabled
				1 = Enabled
	2	IN2R_ENA_STS	0	Input Path 2 (right) enable status
				0 = Disabled
				1 = Enabled
	1	IN1L_ENA_STS	0	Input Path 1 (left) enable status
				0 = Disabled
				1 = Enabled
	0	IN1R_ENA_STS	0	Input Path 1 (right) enable status
				0 = Disabled
				1 = Enabled

Table 4-2. Input Signal Path Enable



4.2.5 Input Signal Path Sample-Rate Control

The input signal paths may be selected as input to the digital mixers or signal-processing functions within the CS42L92 digital core. The sample rate for the input signal paths can be set globally, or can be configured independently for each input channel.

The IN_RATE_MODE bit (defined in Table 4-3) controls whether the input sample rates are set globally using IN_RATE, or independently for each input channel using the INnx_RATE fields (where n is 1–4 and x is L or R for the left/right channels respectively). The IN_RATE and INnx_RATE fields are defined in Table 4-26.

Note that sample-rate conversion is required when routing the input signal paths to any signal chain that is asynchronous or configured for a different sample rate.

4.2.6 Input Signal Path Configuration

The CS42L92 supports up to eight analog inputs or up to eight digital inputs. Selectable combinations of analog (mic or line) and digital inputs are multiplexed into four stereo input signal paths.

Input paths IN1 and IN2 can be configured for single-ended, differential, or DMIC operation. The analog input configuration and pin selection is controlled using the IN*n*x_SRC bits; digital input mode is selected by setting IN*n*_MODE.

Input paths IN3 and IN4 support digital inputs only. Note that the external pin connections are shared with the IN1R and IN2R analog input paths—the following restrictions apply:

- If IN3L or IN3R input paths are enabled, IN1R analog input is restricted to differential (IN1BR–IN1ALN) or single-ended (IN1BR) configurations only.
- If IN4L or IN4R input paths are enabled, IN2R analog input is restricted to single-ended (IN2BR) configuration only.

A configurable high-pass filter (HPF) is provided on the left and right channels of each input path. The applicable cut-off frequency is selected using IN_HPF_CUT. The filter can be enabled on each path independently using the IN*nx*_HPF bits.

The analog input signal paths (single-ended or differential) each incorporate a PGA to provide gain in the range 0 dB to +31 dB in 1-dB steps. Note that these PGAs do not provide pop suppression functions; it is recommended that the gain should not be adjusted while the respective signal path is enabled.

The analog input PGA gain is controlled using IN*n*L_PGA_VOL and IN*n*R_PGA_VOL. Note that separate volume control is provided for the left and right channels of each stereo pair.

If DMIC input is selected, the respective DMICCLKn frequency is controlled by the respective INn_OSR field.

If a signal path is configured for DMIC input, the voltage reference for the associated input/output pins is selectable using the IN*n*_DMIC_SUP fields—each interface may be referenced to MICVDD, MICBIAS1, or MICBIAS2. The voltage reference for each digital input path should be set equal to the applicable power supply of the respective microphones.

The CS42L92 input paths can be configured for power-saving operation, ideal for always-on applications. The low-power configurations allow the power consumption to be optimized with respect to the required audio performance characteristics.

- If a signal path is configured for analog input, low-power operation can be selected by setting the respective INnx_ LP_MODE bit. Mid-power operation can be configured by setting INn_OSR = 100 for the respective signal paths. (Note that the INnx_LP_MODE bit should be cleared in the mid-power configuration.) Note that the maximum input-signal level is reduced by 6 dB if mid-power operation is selected—see Table 3-4.
- If a signal path is configured for digital input, the respective DMICCLK*n* frequency can be configured using the IN*n*_OSR bits. Reducing the DMICCLK*n* frequency reduces power consumption at the expense of audio performance. The IN*n*_OSR field also supports high performance DMIC mode, when 6.144 MHz DMICCLK is selected.
 If 384- or 768-kHz DMICCLK*n* frequency is selected, the maximum sample rate for the respective paths is restricted as described in Table 4-1. If the input sample rates are set globally using IN_RATE (i.e., IN_RATE_MODE = 0), all input paths are affected similarly.

The MICVDD voltage is generated by an internal charge pump and LDO regulator. The MICBIAS*n* outputs are derived from MICVDD; see Section 4.19.



The input signal paths are configured using the fields described in Table 4-3.

Bit	Label	Default	•
10	IN_RATE_	1	Input Path Sample Rate Configuration
	MODE		0 = Global control (all input paths configured using IN_RATE)
			1 = Individual channel control (using the respective INnx_RATE fields)
2:0	IN_HPF_	010	Input Path HPF Select. Controls the cut-off frequency of the input path HPF circuits.
	CUT[2:0]		000 = 2.5 Hz 010 = 10 Hz 100 = 40 Hz
			001 = 5 Hz 011 = 20 Hz All other codes are reserved
15	IN1L_HPF	0	Input Path 1 (Left) HPF Enable
			0 = Disabled
			1 = Enabled
12:11	IN1_DMIC_	00	Input Path 1 DMIC Reference Select (sets the DMICDAT1 and DMICCLK1 logic levels)
	SUP[1:0]		00 = MICVDD 10 = MICBIAS2
			01 = MICBIAS1 11 = Reserved
10	IN1 MODE	0	Input Path 1 Mode
	_		0 = Analog input
			1 = Digital input
7:1	IN1L PGA	0x40	Input Path 1 (Left) PGA Volume (applicable to analog input only)
	VOL[6:0]		0x00 to 0x3F = Reserved 0x42 = 2 dB 0x60 to 0x7F = Reserved
			0x40 = 0 dB (1-dB steps)
			0x41 = 1 dB $0x5F = 31 dB$
14:13	IN1L	00	Input Path 1 (Left) Source
	SRC[1:0]		00 = Differential (IN1ALP–IN1ALN) 10 = Differential (IN1BLP–IN1BLN)
			01 = Single-ended (IN1ALP) 11 = Single-ended (IN1BLP)
11	IN1L LP	0	Input Path 1 (Left) Low-Power Mode (applicable to analog input only)
	MODE		0 = High Performance Mode
			1 = Low Power Mode
10:8	IN1	101	Input Path 1 Oversample Rate Control
	OSR[2:0]		If analog input is selected, this field is used to select Mid-Power Mode.
			100 = Mid Power Mode All other codes are reserved
			101 = Normal
			If digital input is selected, this field controls the DMICCLK1 frequency.
			010 = 384 kHz 101 = 3.072 MHz
			011 = 768 kHz 110 = 6.144 MHz
			100 = 1.536 MHz All other codes are reserved
15	IN1R HPF	0	Input Path 1 (Right) HPF Enable
	_		0 = Disabled
			1 = Enabled
7:1	IN1R PGA	0x40	Input Path 1 (Right) PGA Volume (applicable to analog input only)
	VOL[6:0]		0x00 to 0x3F = Reserved $0x42 = 2 dB$ $0x60 to 0x7F = Reserved$
			0x40 = 0 dB (1-dB steps)
			0x41 = 1 dB $0x5F = 31 dB$
14:13	IN1R	00	Input Path 1 (Right) Source
	SRC[1:0]		00 = Differential (IN1ARP–IN1ARN) 10 = Differential (IN1BR–IN1ALN)
			01 = Single-ended (IN1ARP) 11 = Single-ended (IN1BR)
		<u> </u>	5 ()
11	IIN1R LP	0	Input Path 1 (Right) Low-Power Mode (applicable to analog input only)
11	IN1R_LP_ MODE	0	Input Path 1 (Right) Low-Power Mode (applicable to analog input only) 0 = High Performance Mode
	10 2:0 15 12:11 10 7:1 14:13 11 10:8 15 7:1	10 IN_RATE_MODE 2:0 IN_HPF_CUT[2:0] 15 IN1L_HPF 12:11 IN1_DMIC_SUP[1:0] 10 IN1_MODE 7:1 IN1_PGA_VOL[6:0] 11 IN1L_LPGA_VOL[6:0] 11 IN1L_LP_MODE 10:8 IN1_ODE 15 IN1R_HPF 7:1 IN1L_PGA_VOL[6:0] 11 IN1L_LP_ 10:8 IN1_ODE 15 IN1R_HPF 7:1 IN1R_PGA_VOL[6:0] 14:13 IN1R	10 IN_RATE_ MODE 1 2:0 IN_HPF CUT[2:0] 010 15 IN1L_HPF 0 12:11 IN1_DMIC_ SUP[1:0] 00 10 IN1_MODE 0 7:1 IN1L_PGA_ VOL[6:0] 0x40 14:13 IN1L_LP_GA_ SRC[1:0] 00 11 IN1L_LP_GA_ SRC[1:0] 00 11 IN1L_LP_GA_ SRC[1:0] 010 11 IN1L_LP_GA_ SRC[1:0] 00 11 IN1L_LP_GA_ SRC[1:0] 0 11 IN1L_LP_GA_ SRC[1:0] 0 15 IN1R_ODE 0 7:1 IN1R_HPF 0 7:1 IN1R_PGA_ VOL[6:0] 0x40

Table 4-3. Input Signal Path Configuration



Register Address	Bit	Label	Default	
R792 (0x0318)	15	IN2L_HPF	0	Input Path 2 (Left) HPF Enable
IN2L_Control				0 = Disabled
				1 = Enabled
	12:11	IN2 DMIC	00	Input Path 2 DMIC Reference Select (sets the DMICDAT2 and DMICCLK2 logic levels)
		SUP[1:0]		00 = MICVDD 10 = MICBIAS2
				01 = MICBIAS1 11 = Reserved
	10	IN2 MODE	0	Input Path 2 Mode
			Ū	0 = Analog input
				1 = Digital input
	7:1	IN2L PGA	0x40	Input Path 2 (Left) PGA Volume (applicable to analog input only)
	1.1	VOL[6:0]	0,40	0x00 to 0x3F = Reserved $0x42 = 2 dB$ $0x60 to 0x7F = Reserved$
		10-[0.0]		0x40 = 0 dB (1-dB steps)
D702 (0:0040)	44.40		00	
R793 (0x0319)	14:13	SRC[1:0]	00	Input Path 2 (Left) Source
ADC_Digital_ Volume 2L		010[1.0]		00 = Differential (IN2ALP–IN2ALN) 10 = Differential (IN2BL–IN2ALN)
Volume_ZL	44			01 = Single-ended (IN2ALP) 11 = Single-ended (IN2BL)
	11	IN2L_LP_ MODE	0	Input Path 2 (Left) Low-Power Mode (applicable to analog input only)
		MODE		0 = High Performance Mode
				1 = Low Power Mode
R794 (0x031A)	10:8	IN2	101	Input Path 2 Oversample Rate Control
DMIC2L_Control		OSR[2:0]		If analog input is selected, this field is used to select Mid-Power Mode.
				100 = Mid Power Mode All other codes are reserved
				101 = Normal
				If digital input is selected, this field controls the DMICCLK2 frequency.
				010 = 384 kHz 101 = 3.072 MHz
				011 = 768 kHz 110 = 6.144 MHz
				100 = 1.536 MHz All other codes are reserved
R796 (0x031C)	15	IN2R_HPF	0	Input Path 2 (Right) HPF Enable
IN2R_Control				0 = Disabled
				1 = Enabled
	7:1	IN2R_PGA_	0x40	Input Path 2 (Right) PGA Volume (applicable to analog input only)
		VOL[6:0]		0x00 to 0x3F = Reserved 0x42 = 2 dB 0x60 to 0x7F = Reserved
				0x40 = 0 dB (1-dB steps)
				0x41 = 1 dB 0x5F = 31 dB
R797 (0x0319)	14:13	IN2R	00	Input Path 2 (Right) Source
ADC_Digital_		SRC[1:0]		00 = Differential (IN2ARP–IN2ARN) 10 = Differential (IN2BR–IN2ARN)
Volume_2R				01 = Single-ended (IN2ARP) 11 = Single-ended (IN2BR)
	11	IN2R_LP_	0	Input Path 2 (Right) Low-Power Mode (applicable to analog input only)
		MODE	_	0 = High Performance Mode
				1 = Low Power Mode
R800 (0x0320)	15	IN3L_HPF	0	Input Path 3 (Left) HPF Enable
IN3L_Control			Ŭ	0 = Disabled
				1 = Enabled
	12.11	IN3_DMIC_	00	Input Path 3 DMIC Reference Select (sets the DMICDAT3 and DMICCLK3 logic levels)
		SUP[1:0]	50	00 = MICVDD 10 = MICBIAS2
		- []		01 = MICBIAS1 11 = Reserved
R802 (0x0322)	10:8	IN3	101	Input Path 3 Oversample Rate Control - selects the DMICCLK3 frequency.
DMIC3L_Control	10.0	OSR[2:0]	101	010 = 384 kHz $101 = 3.072 MHz$
		00112.01		
	4 -			100 = 1.536 MHz All other codes are reserved
R804 (0x0324)	15	IN3R_HPF	0	Input Path 3 (Right) HPF Enable
IN3R_Control				
				1 = Enabled

Table 4-3. Input Signal Path Configuration (Cont.)



Register Address	Bit	Label	Default	Description
R808 (0x0328)	15	IN4L_HPF	0	Input Path 4 (Left) HPF Enable
IN4L_Control				0 = Disabled
				1 = Enabled
	12:11	IN4_DMIC_	00	Input Path 4 DMIC Reference Select (sets the DMICDAT4 and DMICCLK4 logic levels)
		SUP[1:0]		00 = MICVDD 10 = MICBIAS2
				01 = MICBIAS1 11 = Reserved
R810 (0x032A)	10:8	—	101	Input Path 4 Oversample Rate Control - selects the DMICCLK4 frequency.
DMIC4L_Control		OSR[2:0]		010 = 384 kHz 101 = 3.072 MHz
				011 = 768 kHz 110 = 6.144 MHz
				100 = 1.536 MHz All other codes are reserved
R812 (0x032C)	15	IN4R_HPF	0	Input Path 4 (Right) HPF Enable
IN4R_Control				0 = Disabled
				1 = Enabled

Table 4-3. Input Signal Path Configuration (Cont.)

4.2.7 Input Signal Path Digital Volume Control

A digital volume control is provided on each input signal path, providing –64 dB to +31.5 dB gain control in 0.5-dB steps. An independent mute control is also provided for each input signal path.

Whenever the gain or mute setting is changed, the signal path gain is ramped up or down to the new settings at a programmable rate. For increasing gain (or unmute), the rate is controlled by IN_VI_RAMP. For decreasing gain (or mute), the rate is controlled by IN_VI_RAMP.

Note: The IN_VI_RAMP and IN_VD_RAMP fields should not be changed while a volume ramp is in progress.

The IN_VU bits control the loading of the input signal path digital volume and mute controls. When IN_VU is cleared, the digital volume and mute settings are loaded into the respective control register, but do not change the signal path gain. The digital volume and mute settings on all of the input signal paths are updated when a 1 is written to IN_VU. This makes it possible to update the gain of multiple signal paths simultaneously.

Note that, although the digital-volume controls provide 0.5-dB steps, the internal circuits provide signal gain adjustment in 0.125-dB steps. This allows a very high degree of gain control and smooth volume ramping under all operating conditions.

Note: The 0 dBFS level of the IN1–IN4 digital input paths is not equal to the 0 dBFS level of the CS42L92 digital core. The maximum digital input signal level is –6 dBFS (see Table 3-7). Under 0 dB gain conditions, a –6 dBFS input signal corresponds to a 0 dBFS input to the CS42L92 digital core functions.

The digital volume control registers are described in Table 4-4 and Table 4-5.

Register Address	Bit	Label	Default	Description				
R777 (0x0309)	6:4	IN_VD_RAMP[2:0]	010	Input Volume Decre	easing Ramp Rate (second	s/6 dB).		
Input_Volume_				This field should no	t be changed while a volur	ne ramp is in progress.		
Ramp				000 = 0 ms	011 = 2 ms	110 = 15 ms		
				001 = 0.5 ms	100 = 4 ms	111 = 30 ms		
				010 = 1 ms	101 = 8 ms			
	2:0	IN_VI_RAMP[2:0]	010	Input Volume Increa	asing Ramp Rate (seconds	i/6 dB).		
				This field should no	t be changed while a volun	ne ramp is in progress.		
				000 = 0 ms	011 = 2 ms	110 = 15 ms		
				001 = 0.5 ms	100 = 4 ms	111 = 30 ms		
				010 = 1 ms	101 = 8 ms			

Table 4-4.	Input Signal	Path Digital	Volume Control
------------	--------------	--------------	----------------



Register Address	Bit	Label	Default	Description
R785 (0x0311) ADC_Digital_	9	IN_VU	See Footnote 1	Input Signal Paths Volume and Mute Update. Writing 1 to this bit causes the Input Signal Paths Volume and Mute settings to be updated simultaneously
Volume_1L	8	IN1L MUTE	1	Input Path 1 (Left) Digital Mute
_		_		0 = Unmute
				1 = Mute
	7:0	IN1L VOL[7:0]	0x80	Input Path 1 (Left) Digital Volume (see Table 4-5 for volume register definition).
				-64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB $0x80 = 0 dB$ $0xC0 to 0xFF = Reserved$
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB
R789 (0x0315)	9	IN_VU	See	Input Signal Paths Volume and Mute Update
ADC Digital		_	Footnote 1	Writing 1 to this bit causes the Input Signal Paths Volume and Mute settings to be
Volume_1R				updated simultaneously
	8	IN1R_MUTE	1	Input Path 1 (Right) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	IN1R_VOL[7:0]	0x80	Input Path 1 (Right) Digital Volume (see Table 4-5 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB 0x80 = 0 dB 0xC0 to 0xFF = Reserved
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB
R793 (0x0319) ADC Digital	9	IN_VU	See Footnote 1	Input Signal Paths Volume and Mute Update. Writing 1 to this bit causes the Input Signal Paths Volume and Mute settings to be updated simultaneously
Volume_2L	8	IN2L_MUTE	1	Input Path 2 (Left) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	IN2L_VOL[7:0]	0x80	Input Path 2 (Left) Digital Volume (see Table 4-5 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB 0x80 = 0 dB 0xC0 to 0xFF = Reserved
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB
R797 (0x031D)	9	IN_VU	See	Input Signal Paths Volume and Mute Update. Writing 1 to this bit causes the Input
ADC_Digital_				Signal Paths Volume and Mute settings to be updated simultaneously
Volume_2R	8	IN2R_MUTE	1	Input Path 2 (Right) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	IN2R_VOL[7:0]	0x80	Input Path 2 (Right) Digital Volume (see Table 4-5 for volume register definition).
				-64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB $0x80 = 0 dB$ $0xC0 to 0xFF = Reserved$
				0x01 = -63.5 dB (0.5-dB steps)
D004 (0, 0004)				(0.5-dB steps) 0xBF = +31.5 dB
R801 (0x0321) ADC_Digital_	9	IN_VU		Input Signal Paths Volume and Mute Update. Writing 1 to this bit causes the Input Signal Paths Volume and Mute settings to be updated simultaneously
Volume_3L	8	IN3L_MUTE	1	Input Path 3 (Left) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	IN3L_VOL[7:0]	0x80	Input Path 3 (Left) Digital Volume (see Table 4-5 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB 0x80 = 0 dB 0xC0 to 0xFF = Reserved
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB

Table 4-4. Input Signal Path Digital Volume Control (Cont.)



Register Address	Bit	Label	Default		Description			
R805 (0x0325)	9	IN_VU	See		Input Signal Paths Volume and Mute Update. Writing 1 to this bit causes the Inpu			
ADC_Digital_			Footnote 1	Signal Paths Volume and Mute settings to be updated simultaneously				
Volume_3R	8	IN3R_MUTE	1	Input Path 3 (Right) Di	igital Mute			
				0 = Unmute				
				1 = Mute				
	7:0	IN3R_VOL[7:0]	0x80		plume (see Table 4-5 for vo	lume register definition).		
				-64 dB to +31.5 dB in	0.5-dB steps			
				0x00 = -64 dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved		
				0x01 = -63.5 dB	(0.5-dB steps)			
				(0.5-dB steps)	0xBF = +31.5 dB			
R809 (0x0329) ADC_Digital_	9	IN_VU	See Footnote 1		ume and Mute Update. Writi and Mute settings to be upd	ing 1 to this bit causes the Input lated simultaneously		
Volume_4L	8	IN4L_MUTE	1	Input Path 4 (Left) Digi	ital Mute			
				0 = Unmute				
				1 = Mute				
	7:0	IN4L_VOL[7:0]	0x80	Input Path 4 (Left) Digi	ital Volume (see Table 4-5 f	for volume register definition).		
				-64 dB to +31.5 dB in	0.5-dB steps			
				0x00 =64 dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved		
				0x01 = -63.5 dB	(0.5-dB steps)			
				(0.5-dB steps)	0xBF = +31.5 dB			
R813 (0x032D) ADC Digital	9	IN_VU	See Footnote 1		ume and Mute Update. Writi and Mute settings to be upd	ing 1 to this bit causes the Input lated simultaneously		
Volume_4R	8	IN4R_MUTE	1	Input Path 4 (Right) Di	igital Mute			
				0 = Unmute				
				1 = Mute				
	7:0	IN4R_VOL[7:0]	0x80	Input Path 4 (Right) Di	igital Volume (see Table 4-5	o for volume register definition).		
				-64 dB to +31.5 dB in	0.5-dB steps			
				0x00 =64 dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved		
				0x01 = -63.5 dB	(0.5-dB steps)			
				(0.5-dB steps)	0xBF = +31.5 dB			

Table 4-4. Input Signal Path Digital Volume Control (Cont.)

1. Default is not applicable to these write-only bits

Table 4-5 lists the input signal path digital volume settings.

Input Volume Register	Volume (dB)						
0x00	-64.0	0x31	-39.5	0x62	-15.0	0x93	9.5
0x01	-63.5	0x32	-39.0	0x63	-14.5	0x94	10.0
0x02	-63.0	0x33	-38.5	0x64	-14.0	0x95	10.5
0x03	-62.5	0x34	-38.0	0x65	-13.5	0x96	11.0
0x04	-62.0	0x35	-37.5	0x66	-13.0	0x97	11.5
0x05	-61.5	0x36	-37.0	0x67	-12.5	0x98	12.0
0x06	-61.0	0x37	-36.5	0x68	-12.0	0x99	12.5
0x07	-60.5	0x38	-36.0	0x69	-11.5	0x9A	13.0
0x08	-60.0	0x39	-35.5	0x6A	-11.0	0x9B	13.5
0x09	-59.5	0x3A	-35.0	0x6B	-10.5	0x9C	14.0
0x0A	-59.0	0x3B	-34.5	0x6C	-10.0	0x9D	14.5
0x0B	-58.5	0x3C	-34.0	0x6D	-9.5	0x9E	15.0
0x0C	-58.0	0x3D	-33.5	0x6E	-9.0	0x9F	15.5
0x0D	-57.5	0x3E	-33.0	0x6F	-8.5	0xA0	16.0
0x0E	-57.0	0x3F	-32.5	0x70	-8.0	0xA1	16.5
0x0F	-56.5	0x40	-32.0	0x71	-7.5	0xA2	17.0
0x10	-56.0	0x41	-31.5	0x72	-7.0	0xA3	17.5
0x11	-55.5	0x42	-31.0	0x73	-6.5	0xA4	18.0

Table 4-5. Input Signal Path Digital Volume Range



Input Volume Register	Volume (dB)						
0x12	-55.0	0x43	-30.5	0x74	-6.0	0xA5	18.5
0x13	-54.5	0x44	-30.0	0x75	-5.5	0xA6	19.0
0x14	-54.0	0x45	-29.5	0x76	-5.0	0xA7	19.5
0x15	-53.5	0x46	-29.0	0x77	-4.5	0xA8	20.0
0x16	-53.0	0x47	-28.5	0x78	-4.0	0xA9	20.5
0x17	-52.5	0x48	-28.0	0x79	-3.5	0xAA	21.0
0x18	-52.0	0x49	-27.5	0x7A	-3.0	0xAB	21.5
0x19	-51.5	0x4A	-27.0	0x7B	-2.5	0xAC	22.0
0x1A	-51.0	0x4B	-26.5	0x7C	-2.0	0xAD	22.5
0x1B	-50.5	0x4C	-26.0	0x7D	-1.5	0xAE	23.0
0x1C	-50.0	0x4D	-25.5	0x7E	-1.0	0xAF	23.5
0x1D	-49.5	0x4E	-25.0	0x7F	-0.5	0xB0	24.0
0x1E	-49.0	0x4F	-24.5	0x80	0.0	0xB1	24.5
0x1F	-48.5	0x50	-24.0	0x81	0.5	0xB2	25.0
0x20	-48.0	0x51	-23.5	0x82	1.0	0xB3	25.5
0x21	-47.5	0x52	-23.0	0x83	1.5	0xB4	26.0
0x22	-47.0	0x53	-22.5	0x84	2.0	0xB5	26.5
0x23	-46.5	0x54	-22.0	0x85	2.5	0xB6	27.0
0x24	-46.0	0x55	-21.5	0x86	3.0	0xB7	27.5
0x25	-45.5	0x56	-21.0	0x87	3.5	0xB8	28.0
0x26	-45.0	0x57	-20.5	0x88	4.0	0xB9	28.5
0x27	-44.5	0x58	-20.0	0x89	4.5	0xBA	29.0
0x28	-44.0	0x59	-19.5	0x8A	5.0	0xBB	29.5
0x29	-43.5	0x5A	-19.0	0x8B	5.5	0xBC	30.0
0x2A	-43.0	0x5B	-18.5	0x8C	6.0	0xBD	30.5
0x2B	-42.5	0x5C	-18.0	0x8D	6.5	0xBE	31.0
0x2C	-42.0	0x5D	-17.5	0x8E	7.0	0xBF	31.5
0x2D	-41.5	0x5E	-17.0	0x8F	7.5	0xC0–0xFF	Reserved
0x2E	-41.0	0x5F	-16.5	0x90	8.0		
0x2F	-40.5	0x60	-16.0	0x91	8.5		
0x30	-40.0	0x61	-15.5	0x92	9.0		

Table 4-5. Input Signal Path Digital Volume Range (Cont.)

4.2.8 Input Signal Path Signal-Detect Control

The CS42L92 provides a digital signal-detect function for the input signal path. This enables system actions to be triggered by signal detection and allows the device to remain in a low-power state until a valid audio signal is detected. A mute function is integrated with the signal-detect circuit, ensuring the respective digital audio path remains at zero until the detection threshold level is reached. Signal detection is also indicated via the interrupt controller.

The signal-detect function is supported on input paths IN1-IN4 in analog and digital configurations. (For input paths IN1 and IN2, digital input is selected by setting the respective INn_MODE bit.) Note that the valid operating conditions for this function vary, depending on the applicable signal-path configuration.

- The signal-detect function is supported on analog input paths for sample rates up to 16 kHz.
- The signal-detect function is supported on digital input paths for sample rates up to 16 kHz (if DMICCLK*n* ≥ 768kHz) and up to 48 kHz (if DMICCLK*n* ≥ 2.8224 MHz).

For each input path, the signal-detect function is enabled by setting the respective IN*nx*_SIG_DET_ENA bit. The detection threshold level is set using IN_SIG_DET_THR—this applies to all input paths.

If the signal-detect function is enabled, the respective input channel is muted if the signal level is below the configured threshold. If the input signal exceeds the threshold level, the respective channel is immediately unmuted.



If the input signal falls below the threshold level, the mute is applied. To prevent erroneous behavior, a time delay is applied before muting the input signal—the channel is only muted if the signal level remains below the threshold level for longer than the hold time. The hold time is set using IN_SIG_DET_HOLD.

Note that the signal-level detection is performed in the digital domain, after the ADC, PGA, digital mute and digital volume controls—the respective input channel must be enabled and unmuted when using the signal-detect function.

The signal-detect function is an input to the interrupt control circuit and can be used to trigger an interrupt event; see Section 4.15. Note that the respective interrupt event represents the logic OR of the signal detection on all input channels and does not provide indication of which input channel caused the interrupt. To avoid multiple interrupts, the signal-detect interrupt can be reasserted only after all input channels have fallen below the trigger threshold level.

The input path signal-detection control registers are described in Table 4-6.

Register Address	Bit	Label	Default		Description	
R786 (0x0312)	15	IN1L_SIG_DET_	0	Input Path 1 (Left) S	ignal-Detect Enable	
DMIC1L Control		ENA		0 = Disabled	•	
-				1 = Enabled		
R790 (0x0316)	15	IN1R_SIG_DET_	0	Input Path 1 (Right)	Signal-Detect Enable	
DMIC1R_Control		ENA		0 = Disabled		
				1 = Enabled		
R794 (0x031A)	15	IN2L_SIG_DET_	0	Input Path 2 (Left) S	ignal-Detect Enable	
DMIC2L_Control		ENA		0 = Disabled		
				1 = Enabled		
R798 (0x031E)	15	IN2R_SIG_DET_	0	Input Path 2 (Right)	Signal-Detect Enable	
DMIC2R_Control		ENA		0 = Disabled		
				1 = Enabled		
R802 (0x0320)	15	IN3L_SIG_DET_	0	Input Path 3 (Left) S	ignal-Detect Enable	
DMIC3L_Control		ENA		0 = Disabled		
				1 = Enabled		
R806 (0x0326)	15	IN3R_SIG_DET_	0		Signal-Detect Enable	
DMIC3R_Control		ENA		0 = Disabled		
				1 = Enabled		
R810 (0x032A)	15	IN4L_SIG_DET_	0	Input Path 4 (Left) S	ignal-Detect Enable	
DMIC4L_Control		ENA		0 = Disabled		
				1 = Enabled		
R814 (0x032E)	15	IN4R_SIG_DET_	0		Signal-Detect Enable	
DMIC4R_Control		ENA		0 = Disabled		
				1 = Enabled		
R832 (0x0340)	8:4	IN_SIG_DET_	0x00		gnal-Detect Threshold	
Signal_Detect_Globals		THR[4:0]		0x00 = -30.1 dB	0x05 = -54.2 dB	0x0A = -72.2 dB
				0x01 = –36.1 dB	0x06 = -56.7 dB	0x0B = -74.7 dB
				0x02 = –42.1 dB	0x07 = –60.2 dB	0x0C = -78.3 dB
				0x03 = –48.2 dB	0x08 = -66.2 dB	0x0D = -80.8 dB
				0x04 = -50.7 dB	0x09 = -68.7 dB	All other codes are reserved
	3:0	IN_SIG_DET_ HOLD[3:0]	0001	Input Signal Path Si is deasserted)	gnal-Detect Hold Time (dela	ay before signal detect indication
				0000 = Reserved	(4-ms steps)	1100 = 96–100 ms
				0001 = 4 - 8 ms	1001 = 36–40 ms	1101 = 192 - 196 ms
				0010 = 8 - 12 ms	1010 = 40-44 ms	1110 = 384–388 ms
				0010 = 0 = 12 ms 0011 = 12 - 16 ms	1010 = 48 - 52 ms	1110 = 768 - 772 ms
						111 100 112 110

Table 4-6. Input Signal Path Signal-Detect Control

4.2.9 Ultrasonic Signal Demodulation

The CS42L92 provides ultrasonic signal-processing functions on the input signal paths. Configurable filters and demodulator functions enable ultrasonic signals to be translated down to the audio band and routed through the digital mixer core. Two ultrasonic processing blocks are incorporated, with independent configuration controls for each.



The input source for the ultrasonic blocks is configured using the US n_SRC fields (where *n* identifies the applicable block US1 or US2). The input signal gain is set using US n_GAIN . The input frequency range is selected by US n_FREQ .

The ultrasonic functions can be supported on any of the input signal paths (IN1–IN4). The inputs to the ultrasonic functions are independent of the enable, mute, and digital-volume settings of the respective input signal paths—the ultrasonic functions do not require the selected input signal paths to be enabled, and the mute/digital-volume settings of the input signal paths have no effect on the ultrasonic functions.

The system clock, SYSCLK, must be present and enabled when using the ultrasonic functions. The SYSCLK frequency must be a multiple of 6.144 MHz (SYSCLK_FRAC = 0) in this case. See Section 4.16 for details of SYSCLK and the associated registers.

The ultrasonic demodulator function is enabled by setting US n_ENA . The frequency band and signal gain are selected using US n_FREQ and US n_GAIN respectively.

The output from the ultrasonic demodulator is a frequency-modulated image of the selected input frequency range. The folding frequency that characterizes the frequency modulation is set according to the US n_FREQ setting—see Table 4-7. The relationship between input and output frequencies is described in Eq. 4-1.

F_{OUT} = |F_{IN} - F_{FOLD}

Equation 4-1. Ultrasonic Demodulator Characteristic

Note that, depending on the input frequency range and the folding frequency, F_{FOLD}, the modulated output in respect of certain input frequencies may overlap others. This effect arises if the folding frequency lies within the input frequency range, with the result that two different input frequencies will each be modulated to the same output frequency. This effect is limited to the outer edges of the input frequency range in all cases. Amplitude response across the input frequency range is flat to within 1.5 dB in all cases.

The demodulated ultrasonic outputs can be selected as input to the digital mixers or signal-processing functions within the digital core by setting the respective x_SRCn fields as described in Section 4.3.1.

The sample rate for the demodulated ultrasonic output is configured using US*n*_RATE—see Table 4-26. The selected sample rate must be one of the SYSCLK-related rates, and must be equal to the output rate set by US*n*_FREQ (see Table 4-7). Note that sample-rate conversion is required when routing the ultrasonic signals to any signal chain that is asynchronous or configured for a different sample rate.

The characteristics associated with the USn_FREQ field setting are shown in Table 4-7.

Condition	Input Frequency Band	Output Sample Rate	Demodulator Folding Frequency (F _{FOLD})
US <i>n_</i> FREQ = 000	24.5–40.5 kHz	32 kHz	40.42 kHz
US <i>n</i> _FREQ = 001	18–22 kHz	8 kHz	18.29 kHz
US <i>n_</i> FREQ = 010	16–24 kHz	16 kHz	16 kHz
US <i>n</i> _FREQ = 011	20–28 kHz	16 kHz	20.21 kHz

Table 4-7. Ultrasonic Frequency Control

The ultrasonic demodulation control registers are described in Table 4-8.



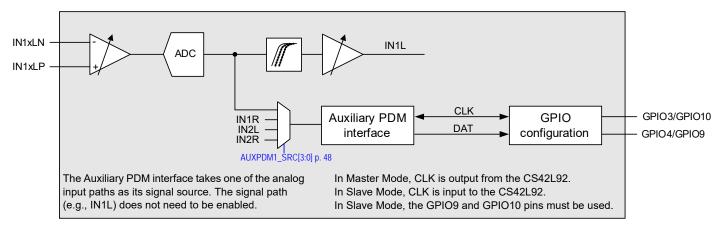
Register Address	Bit	Label	Default		Description	
R4224 (0x1080)	13:12	US1_GAIN[1:0]	10	Ultrasonic Demodulato	r 1 Gain	
US1_Ctrl_0				00 = Disabled (no signa	al) 10 = 1 dB	5
				01 = –5 dB	11 = 7 dB	
	11:8	US1_SRC[3:0]	0x0	Ultrasonic Demodulato	r 1 Source	
				0x0 = IN1L	0x3 = IN2R	0x6 = IN4L
				0x1 = IN1R	0x4 = IN3L	0x7 = IN4R
				0x2 = IN2L	0x5 = IN3R	All other codes are reserved
	6:4	US1_FREQ[2:0]	011	Ultrasonic Demodulato	r 1 Frequency	
				000 = 24.5–40.5 kHz	010 = 16–24 kHz	All other codes are
				001 = 18–22 kHz	011 = 20–28 kHz	reserved
	0	US1_ENA	0	Ultrasonic Demodulato	r 1 Enable	
				0 = Disabled		
				1 = Enabled		
R4226 (0x1082)	13:12 US2_GAIN[1:0]		10	Ultrasonic Demodulato	r 2 Gain	
US2_Ctrl_0				00 = Disabled (no signa	al) 10 = 1 dB	5
				01 = –5 dB	11 = 7 dB	
	11:8	US2_SRC[3:0]	0x0	Ultrasonic Demodulato	r 2 Source	
				0x0 = IN1L	0x3 = IN2R	0x6 = IN4L
				0x1 = IN1R	0x4 = IN3L	0x7 = IN4R
				0x2 = IN2L	0x5 = IN3R	All other codes are reserved
	6:4	US2_FREQ[2:0]	011	Ultrasonic Demodulato	r 2 Frequency	
				000 = 24.5–40.5 kHz	010 = 16–24 kHz	All other codes are
				001 = 18–22 kHz	011 = 20–28 kHz	reserved
	0	US2_ENA	0	Ultrasonic Demodulato	r 2 Enable	
				0 = Disabled		
				1 = Enabled		

Table 4-8. Ultrasonic Signal Demodulation Control

4.2.10 Auxiliary PDM Interface

The auxiliary PDM interface supports a one-channel digital output derived from one of the CS42L92 analog input paths. This can be used to provide an audio path between an analog microphone connected to the CS42L92 and a digital input to an external audio processor.

The auxiliary PDM interface signal path is shown in Fig. 4-13.





The input source for the auxiliary PDM interface is selected using AUXPDM1_SRC. Note that the selected input path must be configured for analog input—the auxiliary PDM function is not supported for the DMICDAT*n* inputs.



Note: The input to the auxiliary PDM interface is independent of the enable, mute, and digital-volume settings of the respective input signal path—the auxiliary PDM function does not require the selected input signal paths to be enabled, and the mute/digital-volume settings of the input signal paths have no effect on the auxiliary PDM function.

The auxiliary PDM interface is enabled by setting AUXPDM1_ENA. Note that the other auxiliary PDM control fields should be configured before enabling the interface. The external connections (GPIO pins) should also be configured before enabling the interface; the AUXPDM1_ENA bit should be set as the final step of the enable sequence. The AUXPDM1_ENA bit should be cleared before changing the interface configuration.

The output signal can be muted and unmuted using AUXPDM1_MUTE.

The interface operates in master or slave modes, selected using the AUXPDM1_MSTR bit. In Master Mode, the clock (CLK) signal is generated by the CS42L92; in Slave Mode, the CLK signal is an input to the CS42L92.

The CLK frequency is selected using the AUXPDM1_CLK_FREQ field. For each setting of this field, the actual frequency depends on whether SYSCLK is configured for 48- or 44.1-kHz related sample rates. See Section 4.16 for details of the system clocks. Note that the CLK frequency must be configured in master and slave modes, using the AUXPDM1_CLK_FREQ field.

The timing of the data (DAT) output signal can be controlled using AUXPDM1_TXEDGE. This selects whether the DAT output changes on the rising or falling edge of CLK.

The auxiliary PDM interface timing is shown in Fig. 4-14. In Master Mode, the clock and data outputs are driven synchronously by the CS42L92. In Slave Mode, the timing of the output data is controlled by the external clock input.

In Slave Mode, the CS42L92 system clock (SYSCLK) must be synchronized to the CLK input. This is achieved by using one of the FLLs to generate the system clock, with AIF2BCLK configured as the FLL clock reference. (If the auxiliary PDM interface is configured in Slave Mode, the CLK input is supported on the AIF2BCLK/GPIO10 pin.) Further details of the FLLs are provided in Section 4.16.8.

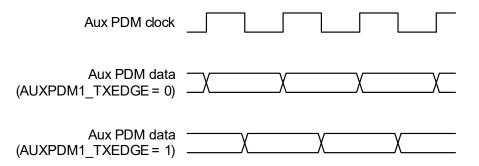


Figure 4-14. Auxiliary PDM Interface Timing

The external connections associated with the auxiliary PDM interface are implemented on GPIO pins, which must be configured for the respective CLK and DAT functions. The auxiliary PDM signals are alternative functions available on specific GPIO pins only. See Section 4.14 to configure the GPIO pins for the auxiliary PDM interface. Note that the output pins (including the CLK function, if Master Mode is selected) must be configured as outputs using the respective GP*n*_DIR fields.

- In Master Mode, the CLK output can be configured on the GPIO3 or GPIO10 pins. The DAT output can be configured on GPIO4 or GPIO9.
- In Slave Mode, the CLK input is supported on GPIO10 only. The DAT output is supported on GPIO9.

The auxiliary PDM interface control registers are described in Table 4-9.



Register Address	Bit	Label	Default	Description	
R4288 (0x10C0)	11:8	AUXPDM1_SRC[3:0]	0x0	Auxiliary PDM 1 Source	
AUXPDM1_Ctrl_0				0x0 = IN1L 0x3 = IN2R	
				0x1 = IN1R All other codes are reserved	
				0x2 = IN2L	
				Note that the selected input source must be configured for analog input— digital input paths are not supported. The Auxiliary PDM interface must be disabled when updating this field.	
	4	AUXPDM1_TXEDGE	0	Auxiliary PDM 1 Timing	
				0 = Output data is driven on rising edge of AUXPDM1_CLK	
		1 = Output data is driven on falling edge of AUXPD			
			The Auxiliary PDM interface must be disabled when updating this field.		
	3	AUXPDM1_MSTR	1	Auxiliary PDM 1 Master Mode select	
				0 = AUXPDM1_CLK Slave Mode (input)	
				1 = AUXPDM1_CLK Master mode (output)	
				The Auxiliary PDM interface must be disabled when updating this field.	
	2	AUXPDM1_MUTE	0	Auxiliary PDM 1 Mute	
				0 = Unmute	
				1 = Mute	
	0	AUXPDM1_ENA	0	Auxiliary PDM 1 Enable	
				0 = Disabled	
				1 = Enabled	
R4281 (0x10C1)	15:14	AUXPDM1_FREQ[1:0]	01	Auxiliary PDM 1 CLK Rate	
AUXPDM1_Ctrl_1				00 = 3.072 MHz (2.8824 MHz) 10 = 1.536 MHz (1.4112 MHz)	
				01 = 2.048 MHz (1.8816 MHz) 11 = 768 kHz (705.6 kHz)	
				The frequencies in brackets apply for 44.1 kHz–related sample rates only (i.e., if SYSCLK_FRAC = 1). The Auxiliary PDM interface must be disabled when updating this field.	

Table 4-9. Auxiliary PDM Interface Control

4.2.11 DMIC Pin Configuration

DMIC operation on input paths IN1–IN4 is selected using IN*n*_MODE, as described in Table 4-3. If DMIC is selected, the respective DMICCLK*n* and DMICDAT*n* pins are configured as digital outputs and inputs, respectively.

The CS42L92 provides integrated pull-down resistors on each DMICDAT*n* pin. This provides a flexible capability for interfacing with other devices.

The DMICDAT*n* pull-down resistors can be configured independently using the bits described in Table 4-10. Note that, if the DMICDAT*n* DMIC input paths are disabled, the pull-down is disabled on the respective pin.

Register Address	Bit	Label	Default	Description
R840 (0x0348)	3	DMICDAT4_PD	0	DMICDAT4 Pull-Down Control
Dig_Mic_Pad_Ctrl				0 = Disabled
				1 = Enabled
	2	DMICDAT3_PD	0	DMICDAT3 Pull-Down Control
				0 = Disabled
				1 = Enabled
	1	DMICDAT2_PD	0	DMICDAT2 Pull-Down Control
				0 = Disabled
				1 = Enabled
	0	DMICDAT1_PD	0	DMICDAT1 Pull-Down Control
				0 = Disabled
				1 = Enabled

Table 4-10. DMIC Interface Pull-Down Control



4.3 Digital Core

The CS42L92 digital core provides extensive mixing and processing capabilities for multiple signal paths. The configuration is highly flexible and supports virtually every conceivable input/output connection between the available processing blocks.

The digital core provides parametric equalization (EQ) functions, DRC, and low-/high-pass filters (LHPF).

The CS42L92 supports multiple signal paths through the digital core. Stereo full-duplex sample-rate conversion is provided to allow digital audio to be routed between input (ADC) paths, output (DAC) paths, digital audio interfaces (AIF1–AIF3) and SLIMbus paths operating at different sample rates or referenced to asynchronous clock domains. Data-format conversion (DFC) functions are available to support different interface standards on the input and output signal paths.

The integrated DSP provides a general-purpose signal processing capability; this is supported by general-purpose timer and event-logger functions. Note that the DSP configuration data is lost whenever the DCVDD power domain is removed; the DSP configuration data must be downloaded to the CS42L92 each time the device is powered up.

The digital core incorporates a S/PDIF transmitter that can provide a stereo S/PDIF output on a GPIO pin. Standard sample rates of 32–192 kHz can be supported. The CS42L92 incorporates a tone generator that can be used for beep functions through any of the audio signal paths. A white-noise generator is incorporated, to provide comfort noise in cases where silence (digital mute) is not desirable.

A haptic signal generator is provided, for use with external haptic devices (e.g., mechanical vibration actuators). Two pulse-width modulation (PWM) signal generators are also provided; the PWM waveforms can be modulated by an audio source within the digital core and can be output on a GPIO pin.

An overview of the digital-core mixing and signal-processing functions is provided in Fig. 4-15. The control registers associated with the digital-core signal paths are shown in Fig. 4-16 through Fig. 4-33. The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The digital audio core is predominantly a 24-bit architecture, but also provides support for 32-bit signal paths. Audio data samples of up to 32 bits can be received via the AIF1, AIF3, and SLIMbus input channels and routed to the AIF1, AIF3, SLIMbus, S/PDIF, and DAC output paths. The respective output mixers provide full support for 32-bit data words.

Note that all other signal paths and signal-processing blocks within the digital core are limited to 24-bit data length; data samples are truncated to 24-bit length if they are routed through any function that does not support 32-bit data words.



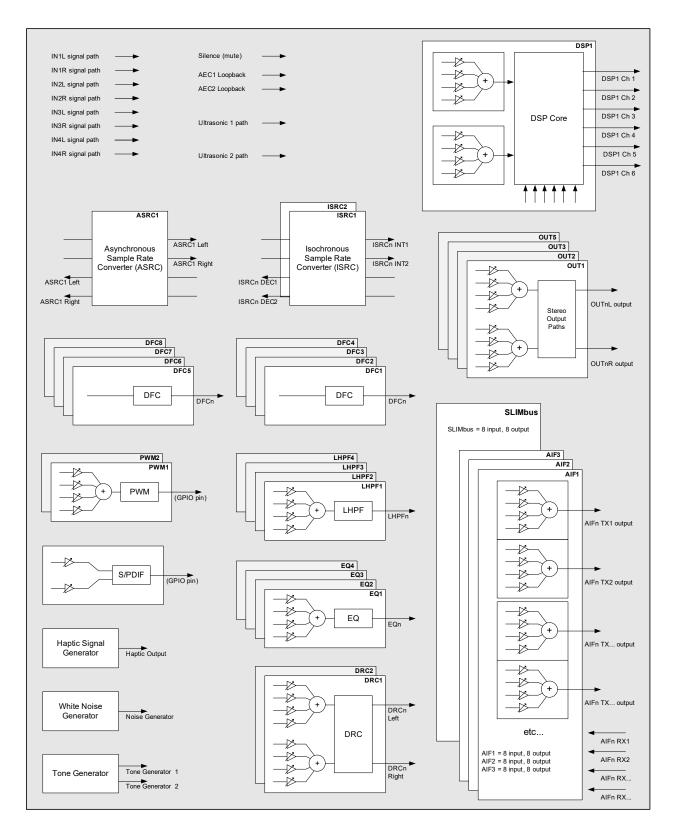


Figure 4-15. Digital Core



4.3.1 Digital-Core Mixers

The CS42L92 provides an extensive digital mixing capability. The digital-core mixing and signal-processing blocks are shown in Fig. 4-15. A four-input digital mixer is associated with many of these functions, as shown. The digital mixer circuit is identical in each instance, providing up to four selectable input sources, with independent volume control on each input.

The control registers associated with the digital-core signal paths are shown in Fig. 4-16–Fig. 4-33. The full list of digital mixer control registers (R1600–R3576) is provided in Section 6.

Further description of the associated control registers is provided throughout Section 4.3. Generic register field definitions are provided in Table 4-11.

The digital mixer input sources are selected using the associated x_SRC*n* fields; the volume control is implemented via the associated x_VOL*n* fields.

The ASRC, ISRC, DFC, and DSP auxiliary input functions support selectable input sources, but do not incorporate any digital mixing. The respective input source (x_SRC*n*) fields are identical to those of the digital mixers.

The x_SRC*n* fields select the input sources for the respective mixer or signal-processing block. Note that the selected input sources must be configured for the same sample rate as the blocks to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16. The DFCs provide support for different data types, including floating point formats. Note that, if unsigned or floating point data is present within the digital core, some restrictions on the valid signal routing options apply—see Section 4.3.13.

A status bit is associated with each configurable input source. If an underclocked error condition occurs, these bits indicate which signal paths have been enabled.

The generic register field definition for the digital mixers is provided in Table 4-11.



Register Address	Bit	Label	Default		Description		
R1600 (0x0640)	15	x_STS <i>n</i>	0	[Digital Core function] input	t <i>n</i> status		
to				0 = Disabled			
R3576 (0x0DF8)				1 = Enabled			
	7:1	x_VOLn	0x40	[Digital Core mixer] input <i>n</i> volume. (–32 dB to +16 dB in 1-dB steps)			
				0x00 to 0x20 = -32 dB	(1-dB steps)	0x50 = +16 dB	
				0x21 = –31 dB	0x40 = 0 dB	0x51 to 0x7F = +16 dB	
				0x22 = -30 dB	(1-dB steps)		
	7:0	x_SRC <i>n</i>	0x00	[Digital Core function] input	t <i>n</i> source select		
				0x00 = Silence (mute)	0x2E = AIF2 RX7	0x63 = LHPF4	
				0x04 = Tone generator 1	0x2F = AIF2 RX8	0x68 = DSP1 Channel 1	
				0x05 = Tone generator 2	0x30 = AIF3 RX1	0x69 = DSP1 Channel 2	
				0x06 = Haptic generator	0x31 = AIF3 RX2	0x6A = DSP1 Channel 3	
				0x08 = AEC Loop-Back 1	0x32 = AIF3 RX3	0x6B = DSP1 Channel 4	
				0x09 = AEC Loop-Back 2	0x33 = AIF3 RX4	0x6C = DSP1 Channel 5	
				0x0D = Noise generator	0x34 = AIF3 RX5	0x6D = DSP1 Channel 6	
				0x10 = IN1L signal path	0x35 = AIF3 RX6	0x90 = ASRC1 IN1 Left	
				0x11 = IN1R signal path	0x36 = AIF3 RX7	0x91 = ASRC1 IN1 Right	
				0x12 = IN2L signal path	0x37 = AIF3 RX8	0x92 = ASRC1 IN2 Left	
				0x13 = IN2R signal path	0x38 = SLIMbus RX1	0x93 = ASRC1 IN2 Right	
				0x14 = IN3L signal path	0x39 = SLIMbus RX2	0xA0 = ISRC1 INT1	
				0x15 = IN3R signal path	0x3A = SLIMbus RX3	0xA1 = ISRC1 INT2	
				0x16 = IN4L signal path	0x3B = SLIMbus RX4	0xA4 = ISRC1 DEC1	
				0x17 = IN4R signal path	0x3C = SLIMbus RX5	0xA5 = ISRC1 DEC2	
				0x20 = AIF1 RX1	0x3D = SLIMbus RX6	0xA8 = ISRC2 INT1	
				0x21 = AIF1 RX2	0x3E = SLIMbus RX7	0xA9 = ISRC2 INT2	
				0x22 = AIF1 RX3	0x3F = SLIMbus RX8	0xAC = ISRC2 DEC1	
				0x23 = AIF1 RX4	0x50 = EQ1	0xAD = ISRC2 DEC2	
				0x24 = AIF1 RX5	0x51 = EQ2	0xF0 = US1	
				0x25 = AIF1 RX6	0x52 = EQ3	0xF1 = US2	
				0x26 = AIF1 RX7	0x53 = EQ4	0xF8 = DFC1	
				0x27 = AIF1 RX8	0x58 = DRC1 Left	0xF9 = DFC2	
				0x28 = AIF2 RX1	0x59 = DRC1 Right	0xFA = DFC3	
				0x29 = AIF2 RX2	0x5A = DRC2 Left	0xFB = DFC4	
				0x2A = AIF2 RX3	0x5B = DRC2 Right	0xFC = DFC5	
				0x2B = AIF2 RX4	0x60 = LHPF1	0xFD = DFC6	
				0x2C = AIF2 RX5	0x61 = LHPF2	0xFE = DFC7	
				0x2D = AIF2 RX6	0x62 = LHPF3	0xFF = DFC8	

Table 4-11. Digital-Core Mixer Control Registers

4.3.2 Digital-Core Inputs

The digital core comprises multiple input paths, as shown in Fig. 4-16. Any of these inputs may be selected as a source to the digital mixers or signal-processing functions within the CS42L92 digital core.

Note that the outputs from other blocks within the digital core may also be selected as input to the digital mixers or signal-processing functions within the CS42L92 digital core. Those input sources, which are not shown in Fig. 4-16, are described separately throughout Section 4.3.

The hexadecimal numbers in Fig. 4-16 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The sample rate for the input signal paths is configured by using the applicable IN_RATE, AIF n_RATE , SLIMRX n_RATE , or US n_RATE field; see Table 4-26. Note that sample-rate conversion is required when routing the input signal paths to any signal chain that is asynchronous or configured for a different sample rate.



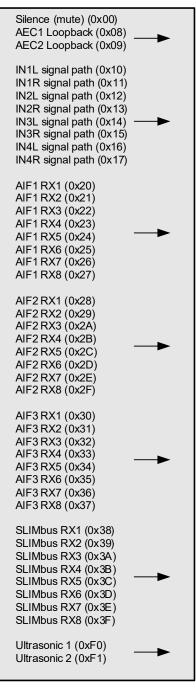


Figure 4-16. Digital-Core Inputs

4.3.3 Digital-Core Output Mixers

The digital core comprises multiple output paths. The output paths associated with AIF1–AIF3 are shown in Fig. 4-17. The output paths associated with OUT1–OUT5 are shown in Fig. 4-18. The output paths associated with the SLIMbus interface are shown in Fig. 4-19.

A four-input mixer is associated with each output. The four input sources are selectable in each case, and independent volume control is provided for each path.



The AIF1–AIF3 output mixer control fields (see Fig. 4-17) are located at register addresses R1792–R1983 (0x0700–0x07BF). The OUT1–OUT5 output mixer control fields (see Fig. 4-18) are located at addresses R1664–R1743 (0x0680–0x06CF). The SLIMbus output mixer control fields (see Fig. 4-19) are located at addresses R1984–R2047 (0x07C0–0x07FF).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the respective mixers. Note that the selected input sources must be configured for the same sample rate as the mixer to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.

The sample rate for the output signal paths is configured using the applicable OUT_RATE, AIF*n*_RATE, or SLIMTX*n*_RATE fields; see Table 4-26. Note that sample-rate conversion is required when routing the output signal paths to any signal chain that is asynchronous or configured for a different sample rate.

The OUT_RATE, AIF*n*_RATE, or SLIMTX*n*_RATE fields must not be changed if any of the respective x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing new values to OUT_RATE, AIF*n*_RATE, or SLIMTX*n*_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to the associated OUT_RATE, AIF*n*_RATE, or SLIMTX*n*_RATE fields. See Table 4-26 for details.

The AIF1, AIF3, SLIMbus, and DAC (OUT1–OUT3) output mixers provide full support for 32-bit data words. Audio data samples up to 32 bits are supported on the AIF1, AIF3, and SLIMbus input channels, which can be routed to the AIF1, AIF3, SLIMbus, and DAC (OUT1–OUT3) output paths. Note that other signal paths and signal-processing blocks within the digital core are limited to 24-bit data length; data samples are truncated to 24-bit length if they are routed through any function that does not support 32-bit data words.

The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the output mixer paths. If the frequency is too low, an attempt to enable an output mixer path fails. Note that active signal paths are not affected under such circumstances.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.



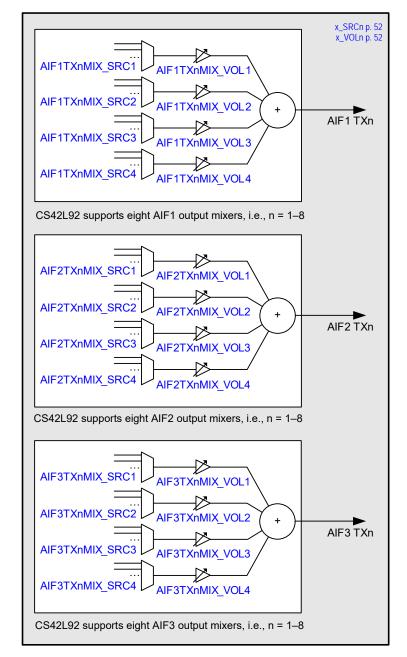


Figure 4-17. Digital-Core AIF Outputs



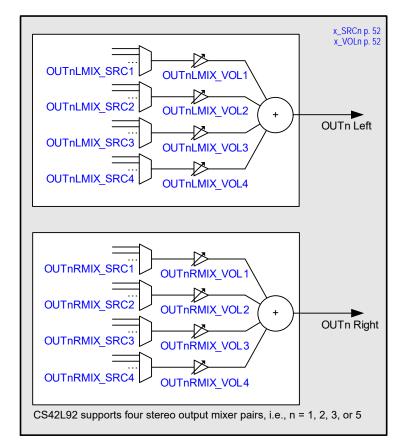


Figure 4-18. Digital-Core OUTn Outputs

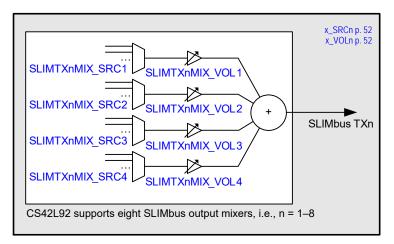


Figure 4-19. Digital-Core SLIMbus Outputs

4.3.4 Five-Band Parametric Equalizer (EQ)

The digital core provides four EQ processing blocks as shown in Fig. 4-20. A four-input mixer is associated with each EQ. The four input sources are selectable in each case, and independent volume control is provided for each path. Each EQ block supports one output.

The EQ provides selective control of five frequency bands as follows:

• The low-frequency band (Band 1) filter can be configured as a peak filter or as a shelving filter. If configured as a shelving filter, it provides adjustable gain below the Band 1 cut-off frequency. As a peak filter, it provides adjustable gain within a defined frequency band that is centered on the Band 1 frequency.



- The midfrequency bands (Band 2–Band 4) filters are peak filters that provide adjustable gain around the respective center frequency.
- The high-frequency band (Band 5) filter is a shelving filter that provides adjustable gain above the Band 5 cut-off frequency.

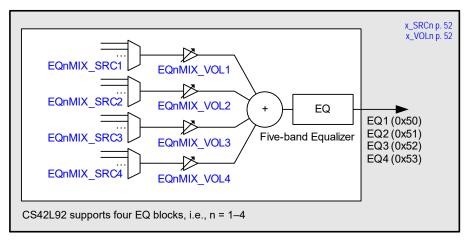


Figure 4-20. Digital-Core EQ Blocks

The EQ1–EQ4 mixer control fields (see Fig. 4-20) are located at register addresses R2176–R2207 (0x0880–0x089F).

The full list of digital-mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the respective EQ processing blocks. Note that the selected input sources must be configured for the same sample rate as the EQ to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.

The hexadecimal numbers in Fig. 4-20 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The sample rate for the EQ function is configured using FX_RATE; see Table 4-26. Note that the EQ, DRC, and LHPF functions must be configured for the same sample rate. Sample-rate conversion is required when routing the EQ signal paths to any signal chain that is asynchronous or configured for a different sample rate.

The FX_RATE field must not be changed if any of the associated x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing a new value to FX_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to FX_RATE. See Table 4-26 for details.

The cut-off or center frequencies for the five-band EQ are set by using the coefficients held in the registers identified in Table 4-12. These coefficients are derived using tools provided in Cirrus Logic's WISCE[™] evaluation-board control software; please contact your Cirrus Logic representative for details.

EQ	Register Addresses
EQ1	R3602 (0x0E10) to R3620 (0x0E24)
EQ2	R3624 (0x0E28) to R3642 (0x0E3A)
EQ3	R3646 (0x0E3E) to R3664 (0x0E53)
EQ4	R3668 (0x0E54) to R3686 (0x0E66)

The control registers associated with the EQ functions are described in Table 4-13.



Register Address	Bit	Label	Default	Description
R3585 (0x0E01)	15:4	FX_STS[11:0]		LHPF, DRC, EQ Enable Status. Indicates the status of each respective
FX_Ctrl2				signal-processing function. Each bit is coded as follows:
_				0 = Disabled
				1 = Enabled
				[11] = EQ4 [7] = DRC2 (Right) [3] = LHPF4
				[10] = EQ3 [6] = DRC2 (Left) [2] = LHPF3
				[9] = EQ2 [5] = DRC1 (Right) [1] = LHPF2
				[8] = EQ1 [4] = DRC1 (Left) [0] = LHPF1
R3600 (0x0E10)		EQ1_B1_GAIN[4:0]		EQ1 Band 1 Gain ¹ (-12 dB to +12 dB in 1-dB steps)
EQ1_1		EQ1_B2_GAIN[4:0]		EQ1 Band 2 Gain 1 (-12 dB to +12 dB in 1-dB steps)
	5:1	EQ1_B3_GAIN[4:0]		EQ1 Band 3 Gain ¹ (-12 dB to +12 dB in 1-dB steps)
	0	EQ1_ENA	0	EQ1 Enable
				0 = Disabled
				1 = Enabled
R3601 (0x0E11)		EQ1_B4_GAIN[4:0]		EQ1 Band 4 Gain 1 (-12 dB to +12 dB in 1-dB steps)
EQ1_2		EQ1_B5_GAIN[4:0]		EQ1 Band 5 Gain ¹ (-12 dB to +12 dB in 1-dB steps)
	0	EQ1_B1_MODE	0	EQ1 Band 1 Mode
				0 = Shelving filter
D0000 (0, 0540) (45.0	504 D4 *		1 = Peak filter
R3602 (0x0E12) to	15:0	EQ1_B1_*	_	EQ1 Frequency Coefficients. Refer to WISCE evaluation board control software for the derivation of these field values.
R3620 (0x0E24)		EQ1_B2_*		
		EQ1_B3_*		
		EQ1_B4_*		
	15:11	EQ1_B5_*	0x0C	EQ2 Band 1 Gain ¹
R3622 (0x0E26)	15:11	EQ2_B1_GAIN[4:0]	UXUC	
EQ2_1	10:6		0x0C	-12 dB to +12 dB in 1-dB steps EQ2 Band 2 Gain ¹
	10.6	EQ2_B2_GAIN[4:0]	UXUC	-12 dB to +12 dB in 1-dB steps
	5:1	EQ2 B3 GAIN[4:0]	0x0C	EQ2 Band 3 Gain ¹
	5.1		0,00	-12 dB to +12 dB in 1-dB steps
	0	EQ2_ENA	0	EQ2 Enable
	Ũ		Ŭ	0 = Disabled
				1 = Enabled
R3623 (0x0E27)	15:11	EQ2_B4_GAIN[4:0]	0x0C	EQ2 Band 4 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
EQ2_2	10:6	EQ2_B5_GAIN[4:0]		EQ2 Band 5 Gain 1 (-12 dB to +12 dB in 1-dB steps)
	0	EQ2_B1_MODE	0	EQ2 Band 1 Mode
				0 = Shelving filter
				1 = Peak filter
R3624 (0x0E28) to	15:0	EQ2_B1_*		EQ2 Frequency Coefficients. Refer to WISCE evaluation board control software for
R3642 (0x0E3A)		EQ2_B2_*		the derivation of these field values.
		EQ2_B3_*		
		EQ2_B4_*		
		EQ2_B5_*		
R3644 (0x0E3C)	15:11	EQ3_B1_GAIN[4:0]	0x0C	EQ3 Band 1 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
EQ3_1	10:6	EQ3_B2_GAIN[4:0]		EQ3 Band 2 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
	5:1	EQ3_B3_GAIN[4:0]	0x0C	EQ3 Band 3 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
	0	EQ3_ENA	0	EQ3 Enable
				0 = Disabled
				1 = Enabled
R3645 (0x0E3D)		EQ3_B4_GAIN[4:0]	0x0C	EQ3 Band 4 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
EQ3_2		EQ3_B5_GAIN[4:0]		EQ3 Band 5 Gain ¹ (-12 dB to +12 dB in 1-dB steps)
	0	EQ3_B1_MODE	0	EQ3 Band 1 Mode
				0 = Shelving filter
				1 = Peak filter

Table 4-13. EQ Enable and Gain Control



Register Address	Bit	Label	Default	
R3646 (0x0E3E) to	15:0	EQ3_B1_*	—	EQ3 Frequency Coefficients. Refer to WISCE evaluation board control software for
R3664 (0x0E50)		EQ3_B2_*		the derivation of these field values.
		EQ3_B3_*		
		EQ3_B4_*		
		EQ3_B5_*		
R3666 (0x0E52)	15:11	EQ4_B1_GAIN[4:0]	0x0C	EQ4 Band 1 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
EQ4_1	10:6	EQ4_B2_GAIN[4:0]	0x0C	EQ4 Band 2 Gain ¹ (-12 dB to +12 dB in 1-dB steps)
	5:1	EQ4_B3_GAIN[4:0]	0x0C	EQ4 Band 3 Gain ¹ (-12 dB to +12 dB in 1-dB steps)
	0	EQ4_ENA	0	EQ4 Enable
				0 = Disabled
				1 = Enabled
R3667 (0x0E53)	15:11	EQ4_B4_GAIN[4:0]	0x0C	EQ4 Band 4 Gain ¹ (–12 dB to +12 dB in 1-dB steps)
EQ4_2	10:6	EQ4_B5_GAIN[4:0]	0x0C	EQ4 Band 5 Gain ¹ (–12 dB to +12 dB in 1-dB steps
	0	EQ4_B1_MODE	0	EQ4 Band 1 Mode
				0 = Shelving filter
				1 = Peak filter
R3668 (0x0E54) to	15:0	EQ4_B1_*		EQ4 Frequency Coefficients. Refer to WISCE evaluation board control software for
R3686 (0x0E66)		EQ4_B2_*		the derivation of these field values.
		EQ4_B3_*		
		EQ4_B4_*		
		EQ4_B5_*		

Table 4-13. EQ Enable and Gain Control (Cont.)

1.See Table 4-14 for gain range.

Table 4-14 lists the EQ gain control settings.

EQ Gain Setting	Gain (dB)	EQ Gain Setting	Gain (dB)
00000	-12	01101	+1
00001	-11	01110	+2
00010	-10	01111	+3
00011	-9	10000	+4
00100	-8	10001	+5
00101	-7	10010	+6
00110	-6	10011	+7
00111	-5	10100	+8
01000	-4	10101	+9
01001	-3	10110	+10
01010	-2	10111	+11
01011	-1	11000	+12
01100	0	11001–11111	Reserved

Table 4-14. EQ Gain-Control Range

The CS42L92 automatically checks to confirm whether the SYSCLK frequency is high enough to support the commanded EQ and digital mixing functions. If an attempt is made to enable an EQ signal path, and there are insufficient SYSCLK cycles to support it, the attempt does not succeed. Note that any signal paths that are already active are not affected under such circumstances.

The FX_STS field in register R3585 indicates the status of each EQ, DRC, and LHPF signal path. If an underclocked error condition occurs, this field indicates which EQ, DRC, or LHPF signal paths have been enabled.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.



4.3.5 Dynamic Range Control (DRC)

The digital core provides two stereo DRC processing blocks, as shown in Fig. 4-21. A four-input mixer is associated with each DRC input channel. The input sources are selectable in each case, and independent volume control is provided for each path. The stereo DRC blocks support two outputs each.

The function of the DRC is to adjust the signal gain in conditions where the input amplitude is unknown or varies over a wide range, for example, when recording from microphones built into a handheld system or to restrict the dynamic range of an output signal path.

To improve intelligibility in the presence of loud impulsive noises, the DRC can apply compression and automatic level control to the signal path. It incorporates anticlip and quick-release features for handling transients.

The DRC also incorporates a noise-gate function that provides additional attenuation of very low-level input signals. This means that the signal path is quiet when no signal is present, giving an improvement in background noise level under these conditions.

A signal-detect function is provided within the DRC; this can be used to detect the presence of an audio signal and to trigger other events. It can also be used as an interrupt event or to trigger the control-write sequencer. Note that DRC triggering of the control-write sequencer is supported for DRC1 only.

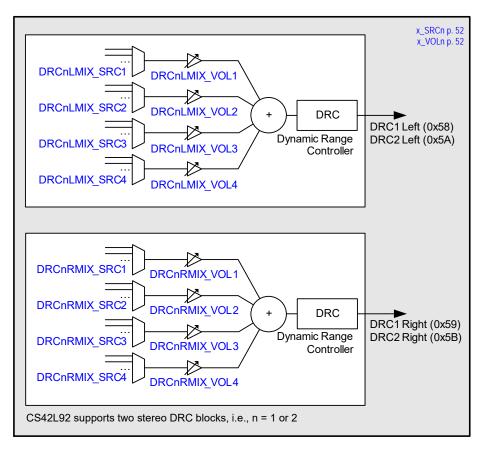


Figure 4-21. Dynamic Range Control (DRC) Block

The DRC1 and DRC2 mixer control fields (see Fig. 4-21) are located at register addresses R2240–R2271 (0x08C0–0x08DF).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the respective DRC processing blocks. Note that the selected input sources must be configured for the same sample rate as the DRC to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.



The hexadecimal numbers in Fig. 4-21 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The sample rate for the DRC function is configured using FX_RATE; see Table 4-26. Note that the EQ, DRC, and LHPF functions must all be configured for the same sample rate. Sample-rate conversion is required when routing the DRC signal paths to any signal chain that is asynchronous or configured for a different sample rate.

The FX_RATE field must not be changed if any of the associated x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing a new value to FX_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to FX_RATE. See Table 4-26 for details.

The DRC functions are enabled using the control registers described in Table 4-15.

Register Address	Bit	Label	Default	Description
R3712 (0x0E80)	1	DRC1L_ENA	0	DRC1 (left) enable
DRC1_ctrl1				0 = Disabled
				1 = Enabled
	0	DRC1R_ENA	0	DRC1 (right) enable
				0 = Disabled
				1 = Enabled
R3720 (0x0E88)	1	DRC2L_ENA	0	DRC2 (left) enable
DRC2_ctrl1				0 = Disabled
				1 = Enabled
	0	DRC2R_ENA	0	DRC2 (right) enable
				0 = Disabled
				1 = Enabled

Table 4-15. DRC Enable

The following description of the DRC is applicable to each DRC. The associated control fields are described in Table 4-17 and Table 4-18 for DRC1 and DRC2 respectively.

4.3.5.1 DRC Compression, Expansion, and Limiting

The DRC supports two different compression regions, separated by a knee at a specific input amplitude. In the region above the knee, the compression slope $DRCn_HI_COMP$ applies; in the region below the knee, the compression slope $DRCn_LO_COMP$ applies. Note that *n* identifies the applicable DRC 1 or 2.

The DRC also supports a noise-gate region, where low-level input signals are heavily attenuated. This function can be enabled or disabled according to the application requirements. The DRC response in this region is defined by the expansion slope DRC*n*_NG_EXP.

For additional attenuation of signals in the noise-gate region, an additional knee can be defined (shown as Knee 2 in Fig. 4-22). When this knee is enabled, this introduces an infinitely steep drop-off in the DRC response pattern between the DRC*n*_LO_COMP and DRC*n*_NG_EXP regions.

The overall DRC compression characteristic in steady state (i.e., where the input amplitude is near constant) is shown in Fig. 4-22.



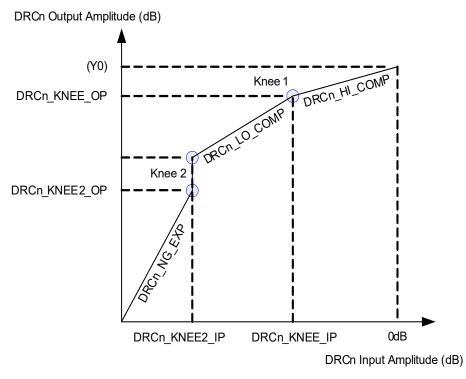


Figure 4-22. DRC Response Characteristic

The slope of the DRC response is determined by DRC*n*_HI_COMP and DRC*n*_LO_COMP. A slope of 1 indicates constant gain in this region. A slope less than 1 represents compression (i.e., a change in input amplitude produces only a smaller change in output amplitude). A slope of 0 indicates that the target output amplitude is the same across a range of input amplitudes; this is infinite compression.

When the noise gate is enabled, the DRC response in this region is determined by DRC*n*_NG_EXP. A slope of 1 indicates constant gain in this region. A slope greater than 1 represents expansion (i.e., a change in input amplitude produces a larger change in output amplitude).

When the DRCn_KNEE2_OP knee is enabled (Knee 2 in Fig. 4-22), this introduces the vertical line in the response pattern shown, resulting in infinitely steep attenuation at this point in the response.

The DRC parameters are listed in Table 4-16.

Parameters	Parameter	Description
1	DRCn_KNEE_IP	Input level at Knee 1 (dB)
2	DRCn_KNEE_OP	Output level at Knee 2 (dB)
3	DRCn_HI_COMP	Compression ratio above Knee 1
4	DRCn_LO_COMP	Compression ratio below Knee 1
5	DRCn_KNEE2_IP	Input level at Knee 2 (dB)
6	DRCn_NG_EXP	Expansion ratio below Knee 2
7	DRCn_KNEE2_OP	Output level at Knee 2 (dB)

Table 4-16.	DRC	Response	Parameters
-------------	-----	----------	------------

The noise gate is enabled by setting DRC*n*_NG_ENA. When the noise gate is not enabled, Parameters 5–7 (see Table 4-16) are ignored, and the DRC*n*_LO_COMP slope applies to all input signal levels below Knee 1.

The DRC*n*_KNEE2_OP knee is enabled by setting DRC*n*_KNEE2_OP_ENA. If this bit is not set, Parameter 7 is ignored and the Knee 2 position always coincides with the low end of the DRC*n*_LO_COMP region.

The Knee 1 point in Fig. 4-22 is determined by DRCn_KNEE_IP and DRCn_KNEE_OP.



Parameter Y0, the output level for a 0 dB input, is not specified directly but can be calculated from the other parameters using Eq. 4-2.

 $Y0 = DRCn_KNEE_OP - (DRCn_KNEE_IP \times DRCn_HI_COMP)$

Equation 4-2. DRC Compression Calculation

4.3.5.2 Gain Limits

The minimum and maximum gain applied by the DRC is set by DRC*n*_MINGAIN, DRC*n*_MAXGAIN, and DRC*n*_NG_MINGAIN. These limits can be used to alter the DRC response from that shown in Fig. 4-22. If the range between maximum and minimum gain is reduced, the extent of the dynamic range control is reduced.

The minimum gain in the compression regions of the DRC response is set by DRC*n*_MINGAIN. The minimum gain in the noise-gate region is set by DRC*n*_NG_MINGAIN. The minimum gain limit prevents excessive attenuation of the signal path.

The maximum gain limit set by DRCn_MAXGAIN prevents quiet signals (or silence) from being excessively amplified.

4.3.5.3 Dynamic Characteristics

The dynamic behavior determines how quickly the DRC responds to changing signal levels. Note that the DRC responds to the average (RMS) signal amplitude over a period of time.

The DRC*n*_ATK determines how quickly the DRC gain decreases when the signal amplitude is high. The DRC*n*_DCY determines how quickly the DRC gain increases when the signal amplitude is low.

These fields are described in Table 4-17 and Table 4-18. The register defaults are suitable for general-purpose microphone use.

4.3.5.4 Anticlip Control

The DRC includes an anticlip function to avoid signal clipping when the input amplitude rises very quickly. This function uses a feed-forward technique for early detection of a rising signal level. Signal clipping is avoided by dynamically increasing the gain attack rate when required.

The anticlip function is enabled using the DRC*n*_ANTICLIP bit. Note that the feed-forward processing increases the latency in the input signal path.

The anticlip feature operates entirely in the digital domain; it cannot be used to prevent signal clipping in the analog domain nor in the source signal. Analog clipping can only be prevented by reducing the analog signal gain or by adjusting the source signal.

It is recommended to disable the anticlip function if the quick-release function (see Section 4.3.5.5) is enabled.

4.3.5.5 Quick Release Control

The DRC includes a quick-release function to handle short transient peaks that are not related to the intended source signal. For example, in handheld microphone recording, transient signal peaks sometimes occur due to user handling, key presses or accidental tapping against the microphone. The quick-release function ensures that these transients do not cause the intended signal to be masked by the longer time constant of DRC*n*_DCY.

The quick-release function is enabled by setting the DRC*n*_QR bit. When this bit is enabled, the DRC measures the crest factor (peak to RMS ratio) of the input signal. A high crest factor is indicative of a transient peak that may not be related to the intended source signal. If the crest factor exceeds the level set by DRC*n*_QR_THR, the normal decay rate (DRC*n*_DCY) is ignored and a faster decay rate (DRC*n*_QR_DCY) is used instead.

It is recommended to disable the quick-release function if the anticlip function (see Section 4.3.5.4) is enabled.



4.3.5.6 Signal Activity Detect

The DRC incorporates a configurable signal-detect function, allowing the signal level at the DRC input to be monitored and to be used to trigger other events. This can be used to detect the presence of a microphone signal on an ADC or DMIC channel or to detect an audio signal received over the digital audio interface.

The DRC signal-detect function is enabled by setting DRC*n*_SIG_DET. Note that the respective DRC*n* must also be enabled. The detection threshold is either a peak level (crest factor) or an RMS level, depending on DRC*n*_SIG_DET_MODE. When peak level is selected, the threshold is determined by DRC*n*_SIG_DET_PK, which defines the applicable crest factor (peak-to-RMS ratio) threshold. If RMS level is selected, the threshold is set using DRC*n*_SIG_DET_RMS.

The DRC signal-detect function is an input to the interrupt control circuit and can be used to trigger an interrupt event; see Section 4.15.

The control-write sequencer can be triggered by the DRC1 signal-detect function. This is enabled by setting DRC1_WSEQ_SIG_DET_ENA. See Section 4.18.

Note that signal detection is supported on DRC1 and DRC2, but the triggering of the control-write sequencer is available on DRC1 only.

4.3.5.7 DRC Register Controls

The DRC1 control registers are described in Table 4-17.

Register Address	Bit	Label	Default	Description				
R3585 (0x0E01) FX_Ctrl2	15:4	FX_STS[11:0]			able status. Indicates the status c unction. Each bit is coded as follo [7] = DRC2 (Right) [6] = DRC2 (Left) [5] = DRC1 (Right) [4] = DRC1 (Left)			

Table 4-17. DRC1 Control Registers



Register Address		Label	Default		Description		
R3712 (0x0E80)	15:11	DRC1_SIG_	0x00			I for signal-detect to be indicated	
DRC1_ctrl1		DET_RMS[4:0]		when DRC1_SIG_DET_MC			
				0x00 = -30 dB	(1.5-dB steps)	0x1F = -76.5 dB	
				0x01 = -31.5 dB	0x1E = -75 dB		
	10:9	IO:9 DRC1_SIG_ 00 DRC1 Signal-Detect Peak Threshold. This is the Peak/RMS ratio, or for signal-detect to be indicated when DRC1_SIG_DET_MODE = 0					
				00 = 12 dB	10 = 24 dB		
				01 = 18 dB	11 = 30 dB		
	8	DRC1_NG_ENA	0	DRC1 Noise-Gate Enable			
			-	0 = Disabled			
				1 = Enabled			
	7	DRC1 SIG	0	DRC1 Signal-Detect Mode			
		DET_MODE		0 = Peak threshold mode			
				1 = RMS threshold mode			
	6	DRC1 SIG DET	0	DRC1 Signal-Detect Enable	9		
				0 = Disabled			
				1 = Enabled			
	5	DRC1_KNEE2_	0	DRC1 KNEE2_OP Enable			
		OP_ENA		0 = Disabled			
				1 = Enabled			
	4	DRC1_QR	1	DRC1 Quick-release Enable	е		
				0 = Disabled			
				1 = Enabled			
-	3	DRC1_ANTICLIP	1	DRC1 Anticlip Enable			
				0 = Disabled			
				1 = Enabled			
	2	DRC1_WSEQ_	0	DRC1 Signal-Detect Write S	Sequencer Select		
		SIG_DET_ENA		0 = Disabled			
				1 = Enabled			
R3713 (0x0E81)	12:9	DRC1_ATK[3:0]	0100	DRC1 Gain attack rate (sec	,		
DRC1_ctrl2				0000 = Reserved	0101 = 2.9 ms	1010 = 92.8 ms	
				0001 = 181 μs	0110 = 5.8 ms	1011 = 185.6 ms	
				0010 = 363 μs	0111 = 11.6 ms	1100 to 1111 = Reserved	
				0011 = 726 μs	1000 = 23.2 ms		
	0.5		4004	0100 = 1.45 ms	1001 = 46.4 ms		
	8:5	DRC1_DCY[3:0]	1001	DRC1 Gain decay rate (sec	-	1010 1 10	
				0000 = 1.45 ms	0101 = 46.5 ms	1010 = 1.49 s	
				0001 = 2.9 ms	0110 = 93 ms	1011 = 2.97 s	
				0010 = 5.8 ms	0111 = 186 ms	1100 to 1111 = Reserved	
				0011 = 11.6 ms	1000 = 372 ms		
	4.0	DBC1	100	0100 = 23.25 ms	1001 = 743 ms		
	4:2	DRC1_ MINGAIN[2:0]	100	DRC1 Minimum gain to atte	_	11V - Popprind	
				000 = 0 dB 001 = –12 dB	011 = –24 dB 100 = –36 dB	11X = Reserved	
	1:0	DRC1	11	010 = –18 dB DRC1 Maximum gain to boo	101 = Reserved		
	1.0	MAXGAIN[1:0]		00 = 12 dB	10 = 24 dB		
				00 – 12 dB 01 = 18 dB	10 – 24 dB 11 = 36 dB		
					11 – 30 UD		

Table 4-17. DRC1 Control Registers (Cont.)



Register Address	Bit	Label	Default		Description			
R3714 (0x0E82)	15:12	DRC1_NG_	0000	DRC1 Minimum gain to atte	enuate audio signals when the	e Noise Gate is active.		
DRC1_ctrl3		MINGAIN[3:0]		0000 = -36 dB	0101 = –6 dB	1010 = 24 dB		
				0001 = -30 dB	0110 = 0 dB	1011 = 30 dB		
				0010 = -24 dB	0111 = 6 dB	1100 = 36 dB		
				0011 = -18 dB	1000 = 12 dB	1101 to 1111 = Reserved		
				0100 = –12 dB	1001 = 18 dB			
	11:10	DRC1_NG_	00	DRC1 Noise-Gate slope				
		EXP[1:0]		00 = 1 (no expansion)	10 = 4			
				01 = 2	11 = 8			
	9:8	DRC1_QR_	00	DRC1 Quick-release thresh	old (crest factor in dB)			
		THR[1:0]		00 = 12 dB	10 = 24 dB			
				01 = 18 dB	11 = 30 dB			
	7:6	DRC1_QR_	00	DRC1 Quick-release decay	rate (seconds/6 dB)			
		DCY[1:0]		00 = 0.725 ms	10 = 5.8 ms			
				01 = 1.45 ms	11 = Reserved			
	5:3	DRC1_HI_	011	DRC1 Compressor slope (u	upper region)			
		COMP[2:0]		000 = 1 (no compression)	011 = 1/8	11X = Reserved		
				001 = 1/2	100 = 1/16			
				010 = 1/4	101 = 0			
	2:0	DRC1_LO_	000	DRC1 Compressor slope (I	3,			
		COMP[2:0]		000 = 1 (no compression)	011 = 1/8	11X = Reserved		
				001 = 1/2	100 = 0			
				010 = 1/4	101 = Reserved			
R3715 (0x0E83)	10:5	DRC1_KNEE_	0x00	DRC1 Input signal level at t	the compressor knee.			
DRC1_ctrl4		IP[5:0]		0x00 = 0 dB	0x02 = -1.5 dB	0x3C = -45 dB		
				0x01 = -0.75 dB	(–0.75-dB steps)	0x3D–0x3F = Reserved		
	4:0	DRC1_KNEE_	0x00	DRC1 Output signal at the				
		OP[4:0]		0x00 = 0 dB	0x02 = -1.5 dB	0x1E = -22.5 dB		
				0x01 = -0.75 dB	(–0.75 dB steps)	0x1F = Reserved		
R3716 (0x0E84)	9:5	DRC1_KNEE2_	0x00	DRC1 Input signal level at t	the noise-gate threshold, Kne	e 2.		
DRC1_ctrl5		IP[4:0]		0x00 = -36 dB	0x02 = -39 dB	0x1E = -81 dB		
				0x01 = -37.5 dB	(-1.5-dB steps)	0x1F = -82.5 dB		
				Applicable if DRC1_NG_ENA = 1				
	4:0	DRC1_KNEE2_	0x00		noise-gate threshold, Knee 2			
		OP[4:0]		0x00 = -30 dB	0x02 = -33 dB	0x1E = -75 dB		
				0x01 = -31.5 dB	(–1.5dB steps)	0x1F = -76.5 dB		
				Applicable only if DRC1_K	NEE2_OP_ENA = 1			

Table 4-17. DRC1 Control Registers (Cont.)

The DRC2 control registers are described in Table 4-18.

Table 4-18.	DRC2	Control	Registers
-------------	------	---------	-----------

Register Address	Bit	Label	Default	Description		
R3585 (0x0E01) FX Ctrl2	15:4	FX_STS[11:0]		LHPF, DRC, EQ Enable Status. Indicates the status of each respective signal-processing function. Each bit is coded as follows:		•
				0 = Disabled		
				1 = Enabled		
				[11] = EQ4	[7] = DRC2 (Right)	[3] = LHPF4
				[10] = EQ3	[6] = DRC2 (Left)	[2] = LHPF3
				[9] = EQ2	[5] = DRC1 (Right)	[1] = LHPF2
				[8] = EQ1	[4] = DRC1 (Left)	[0] = LHPF1



Register Address		Label	Default	Descriptio	n
R3720 (0x0E88)	15:11	DRC2_SIG_	0x00	DRC2 Signal-Detect RMS Threshold. This is the	
DRC2_ctrl1		DET_RMS[4:0]		be indicated when DRC2_SIG_DET_MODE = 1	
				0x00 = -30 dB (1.5-dB steps)	0x1E = -75 dB
				0x01 = -31.5 dB	0x1F = -76.5 dB
	10:9	DRC2_SIG_ DET_PK[1:0]	00	DRC2 Signal-Detect Peak Threshold. Peak/RM signal-detect to be indicated when DRC2_SIG_	S ratio, or Crest Factor, level for DET_MODE = 0.
				00 = 12 dB 10 = 24 dB	
				01 = 18 dB 11 = 30 dB	
	8	DRC2_NG_ENA	0	DRC2 Noise-Gate Enable	
				0 = Disabled	
				1 = Enabled	
	7	DRC2_SIG_	0	DRC2 Signal-Detect Mode	
		DET_MODE		0 = Peak threshold mode	
				1 = RMS threshold mode	
	6	DRC2_SIG_DET	0	DRC2 Signal-Detect Enable	
				0 = Disabled	
				1 = Enabled	
	5	DRC2_KNEE2_	0	DRC2 KNEE2_OP Enable	
		OP_ENA		0 = Disabled	
				1 = Enabled	
	4	DRC2_QR	1	DRC2 Quick-release Enable	
				0 = Disabled	
	•			1 = Enabled	
	3	DRC2_ANTICLIP	1	DRC2 Anticlip Enable	
				0 = Disabled	
	10.0	DRC2 ATK[3:0]	0100	1 = Enabled	
R3721 (0x0E89)	12:9	DRC2_AIK[3:0]	0100	DRC2 Gain attack rate (seconds/6 dB) 0000 = Reserved 0101 = 2.9 ms	1010 - 02.0 ma
DRC2_ctrl2					1010 = 92.8 ms 1011 = 185.6 ms
				0001 = 181 μs0110 = 5.8 ms0010 = 363 μs0111 = 11.6 ms	1001 to 1111 = Reserved
				0010 = 363 μs0111 = 11.6 ms0011 = 726 μs1000 = 23.2 ms	1100 to 1111 – Reserved
				0100 = 1.45 ms $1000 = 23.2 ms1001 = 46.4 ms$	
	8:5	DRC2_DCY[3:0]	1001	DRC2 Gain decay rate (seconds/6 dB)	
	0.0	DIX02_D01[0.0]	1001	0000 = 1.45 ms $0101 = 46.5 ms$	1010 = 1.49 s
				0001 = 2.9 ms 0110 = 93 ms	1010 = 1.40 S
				0010 = 5.8 ms 0111 = 186 ms	1100 to 1111 = Reserved
				0011 = 11.6 ms 1000 = 372 ms	
				0100 = 23.25 ms 1001 = 743 ms	
	4:2	DRC2_	100	DRC2 Minimum gain to attenuate audio signals	
		MINGAIN[2:0]		000 = 0 dB $011 = -24 dB$	11X = Reserved
				001 = -12 dB (default) $100 = -36 dB$	
				010 = -18 dB $101 = Reserved$	
	1:0	DRC2	11	DRC2 Maximum gain to boost audio signals (dE	3)
		MAXGAIN[1:0]		00 = 12 dB $10 = 24 dB$,
				01 = 18 dB 11 = 36 dB	

Table 4-18. DRC2 Control Registers (Cont.)



Register Address	Bit	Label	Default		Description	
R3722 (0x0E8A)	15:12	DRC2_NG_	0000	DRC2 Minimum gain to atte	enuate audio signals when th	e Noise Gate is active.
DRC2_ctrl3		MINGAIN[3:0]		0000 = -36 dB	0101 = –6 dB	1010 = 24 dB
				0001 = -30 dB	0110 = 0 dB	1011 = 30 dB
				0010 = –24 dB	0111 = 6 dB	1100 = 36 dB
				0011 = –18 dB	1000 = 12 dB	1101 to 1111 = Reserved
				0100 = –12 dB	1001 = 18 dB	
	11:10	DRC2_NG_	00	DRC2 Noise-Gate slope		
		EXP[1:0]		00 = 1 (no expansion)	10 = 4	
				01 = 2	11 = 8	
	9:8	DRC2_QR_	00	DRC2 Quick-release thresh	nold (crest factor in dB)	
		THR[1:0]		00 = 12 dB	10 = 24 dB	
				01 = 18 dB	11 = 30 dB	
	7:6	DRC2_QR_	00	DRC2 Quick-release decay	rate (seconds/6 dB)	
		DCY[1:0]		00 = 0.725 ms	10 = 5.8 ms	
				01 = 1.45 ms	11 = Reserved	
	5:3	DRC2_HI_	011	DRC2 Compressor slope (u	upper region)	
		COMP[2:0]		000 = 1 (no compression)	011 = 1/8	11X = Reserved
				001 = 1/2	100 = 1/16	
				010 = 1/4	101 = 0	
	2:0	DRC2_LO_	000	DRC2 Compressor slope (I	ower region)	
		COMP[2:0]		000 = 1 (no compression)	011 = 1/8	11X = Reserved
				001 = 1/2	100 = 0	
				010 = 1/4	101 = Reserved	
R3723 (0x0E8B)	10:5	DRC2_KNEE_	000000	DRC2 Input signal level at t	the compressor knee.	
DRC2_ctrl4		IP[5:0]		0x00 = 0 dB	0x02 = -1.5 dB	0x3C = -45 dB
				0x01 = –0.75 dB	(–0.75-dB steps)	0x3D–0x3F = Reserved
	4:0	DRC2_KNEE_	00000	DRC2 Output signal at the	compressor knee.	
		OP[4:0]		0x00 = 0 dB	0x02 = -1.5 dB	0x1E = -22.5 dB
				0x01 = –0.75 dB	(–0.75 dB steps)	0x1F = Reserved
R3724 (0x0E8C)	9:5	DRC2_KNEE2_	00000	DRC2 Input signal level at t	the noise-gate threshold, Kne	ee 2.
DRC2_ctrl5		IP[4:0]		0x00 = -36 dB	0x02 = -39 dB	0x1E =81 dB
				0x01 = -37.5 dB	(-1.5-dB steps)	0x1F = -82.5 dB
				Applicable only if DRC2_N		
	4:0	DRC2_KNEE2_	00000	DRC2 Output signal at the	noise-gate threshold, Knee 2	
		OP[4:0]		0x00 = -30 dB	0x02 = -33 dB	0x1E = -75 dB
				0x01 = -31.5 dB	(–1.5dB steps)	0x1F = -76.5 dB
				Applicable only if DRC2_K	NEE2_OP_ENA = 1.	

Table 4-18. DRC2 Control Registers (Co.	nt.)
---	------

The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the commanded DRC and digital mixing functions. If the frequency is too low, an attempt to enable a DRC signal path fails. Note that active signal paths are not affected under such circumstances.

The FX_STS field in register R3585 indicates the status of each EQ, DRC, and LHPF signal path. If an underclocked error condition occurs, this field indicates which EQ, DRC, or LHPF signal paths have been enabled.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

4.3.6 Low-/High-Pass Digital Filter (LHPF)

The digital core provides four LHPF processing blocks as shown in Fig. 4-23. A four-input mixer is associated with each filter. The four input sources are selectable in each case, and independent volume control is provided for each path. Each LHPF block supports one output.

The LHPF /HPF can be used to remove unwanted out-of-band noise from a signal path. Each filter can be configured either as a low-pass filter (LPF) or a high-pass filter (HPF).



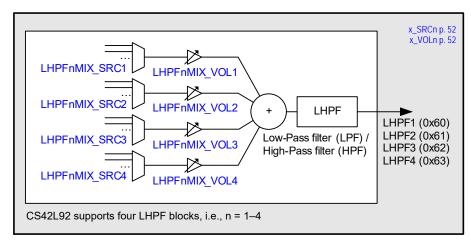


Figure 4-23. Digital-Core LPF/HPF Blocks

The LHPF1–LHPF4 mixer control fields, shown in Fig. 4-23, are located at register addresses R2304–R2335 (0x0900–0x091F).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the respective LHPF processing blocks. Note that the selected input sources must be configured for the same sample rate as the LHPF to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.

The hexadecimal numbers in Fig. 4-23 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The sample rate for the LHPF function is configured using FX_RATE; see Table 4-26. Note that the EQ, DRC, and LHPF functions must all be configured for the same sample rate. Sample-rate conversion is required when routing the LHPF signal paths to any signal chain that is asynchronous or configured for a different sample rate.

The FX_RATE field must not be changed if any of the associated x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing a new value to FX_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to FX_RATE. See Table 4-26 for details.

The control registers associated with the LHPF functions are described in Table 4-19.

The cut-off frequencies for the LHPF blocks are set by using the coefficients held in registers R3777, R3781, R3785, and R3789 for LHPF1, LHPF2, LHPF3 and LHPF4 respectively. These coefficients are derived using tools provided in Cirrus Logic's WISCE evaluation board control software; please contact your Cirrus Logic representative for details.

Register Address	Bit	Label	Default		Description	
R3585 (0x0E01) FX Ctrl2	15:4	FX_STS[11:0]		· · · ·	able Status. Indicates the status unctions. Each bit is coded as fo	
-				0 = Disabled		
				1 = Enabled		
				[11] = EQ4	[7] = DRC2 (Right)	[3] = LHPF4
				[10] = EQ3	[6] = DRC2 (Left)	[2] = LHPF3
				[9] = EQ2	[5] = DRC1 (Right)	[1] = LHPF2
				[8] = EQ1	[4] = DRC1 (Left)	[0] = LHPF1



Register Address		Label	Default	
R3776 (0x0EC0)	1	LHPF1_MODE	0	Low-/High-Pass Filter 1 Mode
HPLPF1_1				0 = Low Pass
				1 = High Pass
	0	LHPF1_ENA	0	Low-/High-Pass Filter 1 Enable
				0 = Disabled
				1 = Enabled
R3777 (0x0EC1)	15:0	LHPF1_COEFF[15:0]	0x0000	Low-/High-Pass Filter 1 Frequency Coefficient
HPLPF1_2				Refer to WISCE evaluation board control software for the derivation of this field
				value.
R3780 (0x0EC4)	1	LHPF2_MODE	0	Low-/High-Pass Filter 2 Mode
HPLPF2_1				0 = Low Pass
				1 = High Pass
	0	LHPF2_ENA	0	Low-/High-Pass Filter 2 Enable
				0 = Disabled
				1 = Enabled
R3781 (0x0EC5)	15:0	LHPF2_COEFF[15:0]	0x0000	Low-/High-Pass Filter 2 Frequency Coefficient
HPLPF2_2				Refer to WISCE evaluation board control software for the derivation of this field
				value.
R3784 (0x0EC8)	1	LHPF3_MODE	0	Low-/High-Pass Filter 3 Mode
HPLPF3_1				0 = Low Pass
				1 = High Pass
	0	LHPF3_ENA	0	Low-/High-Pass Filter 3 Enable
				0 = Disabled
				1 = Enabled
R3785 (0x0EC9)	15:0	LHPF3_COEFF[15:0]	0x0000	Low-/High-Pass Filter 3 Frequency Coefficient
HPLPF3_2				Refer to WISCE evaluation board control software for the derivation of this field
				value.
R3788 (0x0ECC)	1	LHPF4_MODE	0	Low-/High-Pass Filter 4 Mode
HPLPF4_1				0 = Low Pass
				1 = High Pass
	0	LHPF4_ENA	0	Low-/High-Pass Filter 4 Enable
				0 = Disabled
				1 = Enabled
R3789 (0x0ECD)	15:0	LHPF4_COEFF[15:0]	0x0000	Low-/High-Pass Filter 4 Frequency Coefficient
HPLPF4_2				Refer to WISCE evaluation board control software for the derivation of this field
				value.

Table 4-19. Low-Pass Filter/High-Pass Filter (Cont.)

The CS42L92 performs automatic checks to confirm whether the SYSCLK frequency is high enough to support the commanded LHPF and digital mixing functions. If the frequency is too low, an attempt to enable an LHPF signal path fails. Note that active signal paths are not affected under such circumstances.

The FX_STS field in register R3585 indicates the status of each EQ, DRC, and LHPF signal path. If an underclocked error condition occurs, this field indicates which EQ, DRC, or LHPF signal paths have been enabled.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

4.3.7 Digital-Core DSP

The digital core provides one programmable DSP processing block as shown in Fig. 4-24. The DSP block supports eight inputs (Left, Right, Aux1, Aux2, ... Aux6). A four-input mixer is associated with the left and right inputs, providing further expansion of the number of input paths. Each of the input sources is selectable, and independent volume control is provided for left and right input mixer channels. The DSP block supports six outputs.

The functionality of the DSP processing block is not fixed; application-specific algorithms can be implemented according to different customer requirements. The procedure for configuring the CS42L92 DSP functions is tailored to each customer's application; please contact your Cirrus Logic representative for details.



For details of the DSP firmware requirements relating to clocking, register access, and code execution, refer to Section 4.4.3.

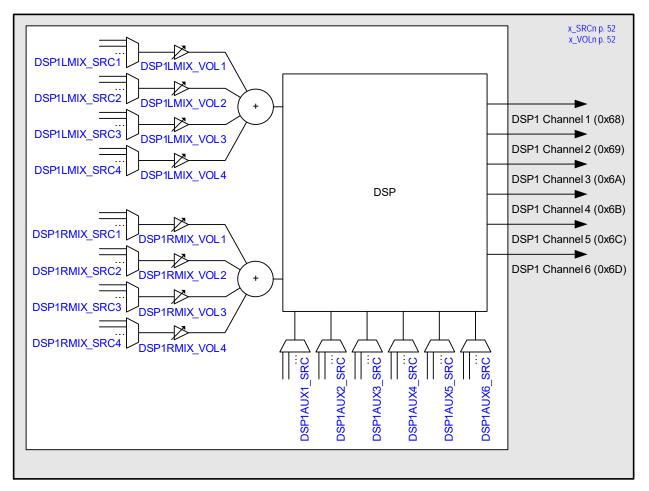


Figure 4-24. Digital-Core DSP Block

The DSP mixer input control fields (see Fig. 4-24) are located at register addresses R2368–R2424 (0x0940–0x0978).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the DSP processing block. Note that the selected input sources must be configured for the same sample rate as the DSP. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.16.

The hexadecimal numbers in Fig. 4-24 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The sample rate for the DSP functions is configured using the DSP1_RATE field; see Table 4-26. Sample-rate conversion is required when routing the DSP signal paths to any signal chain that is configured for a different sample rate.

The DSP1_RATE field must not be changed if any of the respective x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing new values to DSP1_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to the DSP1_RATE field. See Table 4-26 for details.

The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the required DSP mixing functions. If the frequency is too low, an attempt to enable a DSP mixer path fails. Note that active signal paths are not affected under such circumstances.



The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

4.3.8 S/PDIF Output Generator

The CS42L92 incorporates an IEC-60958-3–compatible S/PDIF output generator, as shown in Fig. 4-25; this provides a stereo S/PDIF output on a GPIO pin. The S/PDIF transmitter allows full control over the S/PDIF validity bits and channel status information.

The input sources to the S/PDIF transmitter are selectable for each channel, and independent volume control is provided for each path. The *TX1 and *TX2 fields control Channels A and B (respectively) of the S/PDIF output.

The S/PDIF signal can be output directly on a GPIO pin. See Section 4.14 to configure a GPIO pin for this function.

Note that the S/PDIF signal cannot be selected as input to the digital mixers or signal-processing functions within the CS42L92 digital core.

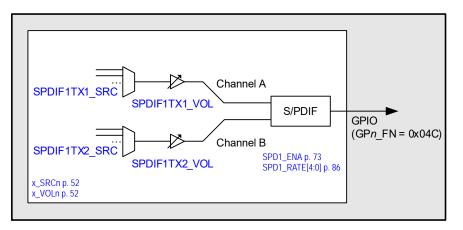


Figure 4-25. Digital-Core S/PDIF Output Generator

The S/PDIF input control fields (see Fig. 4-25) are located at register addresses R2048–R2057 (0x0800–0x0809).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the two S/PDIF channels. Note that the selected input sources must be synchronized to the SYSCLK clocking domain, and configured for the same sample rate as the S/PDIF generator. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.

The sample rate of the S/PDIF generator is configured using SPD1_RATE; see Table 4-26. The S/PDIF transmitter supports sample rates in the range 32–192 kHz. Note that sample-rate conversion is required when linking the S/PDIF generator to any signal chain that is asynchronous or configured for a different sample rate.

The SPD1_RATE field must not be changed if any of the associated x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing a new value to SPD1_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to SPD1_RATE. See Table 4-26 for details.

The S/PDIF generator is enabled by setting SPD1_ENA, as described in Table 4-20.

The S/PDIF output contains audio data derived from the selected sources. Audio samples up to 24-bit width can be accommodated. The validity bits and the channel status bits in the S/PDIF data are configured using the corresponding fields in registers R1474 (0x5C2) to R1477 (0x5C5).

Refer to the S/PDIF specification (IEC60958-3 Digital Audio Interface - Consumer) for full details of the S/PDIF protocol and configuration parameters.



Register Address	Bit	Label	Default	Description
R1474 (0x05C2)	13	SPD1_VAL2	0	S/PDIF Validity (Subframe B)
SPD1_TX_Control	12	SPD1_VAL1	0	S/PDIF Validity (Subframe A)
	0	SPD1_ENA	0	S/PDIF Generator Enable
				0 = Disabled
				1 = Enabled
R1475 (0x05C3)	15:8	SPD1_CATCODE[7:0]	0x00	S/PDIF Category code
SPD1_TX_	7:6	SPD1_CHSTMODE[1:0]	00	S/PDIF Channel Status mode
Channel_Status_1	5:3	SPD1_PREEMPH[2:0]	000	S/PDIF Preemphasis mode
	2	SPD1_NOCOPY	0	S/PDIF Copyright status
	1	SPD1_NOAUDIO	0	S/PDIF Audio/nonaudio indication
	0	SPD1_PRO	0	S/PDIF Consumer Mode/Professional Mode
R1476 (0x05C4)	15:12	SPD1_FREQ[3:0]	0000	S/PDIF Indicated sample frequency
SPD1_TX_	11:8	SPD1_CHNUM2[3:0]	1011	S/PDIF Channel number (Subframe B)
Channel_Status_2	7:4	SPD1_CHNUM1[3:0]	0000	S/PDIF Channel number (Subframe A)
	3:0	SPD1_SRCNUM[3:0]	0001	S/PDIF Source number
R1477 (0x05C5)	11:8	SPD1_ORGSAMP[3:0]	0000	S/PDIF Original sample frequency
SPD1_TX_	7:5	SPD1_TXWL[2:0]	000	S/PDIF Audio sample word length
Channel_Status_3	4	SPD1_MAXWL	0	S/PDIF Maximum audio sample word length
	3:2	SPD1_SC31_30[1:0]	00	S/PDIF Channel Status [31:30]
	1:0	SPD1_CLKACU[1:0]	00	Transmitted Clock accuracy

Table 4-20. S/PDIF Output Generator Control

The S/PDIF output generator provides full support for 32-bit data words. Audio data samples up to 32 bits are supported on the AIF1, AIF3, and SLIMbus input channels, which can be routed to the S/PDIF output. Note that other signal paths and signal-processing blocks within the digital core are limited to 24-bit data length; data samples are truncated to 24-bit length if they are routed through any function that does not support 32-bit data words.

The CS42L92 automatically checks to confirm whether the SYSCLK frequency is high enough to support the digital mixer paths. If an attempt is made to enable the S/PDIF generator, and there are insufficient SYSCLK cycles to support it, the attempt does not succeed. Note that any active signal paths are unaffected under such circumstances.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

4.3.9 Tone Generator

The CS42L92 incorporates a tone generator that can be used for beep functions through any of the audio signal paths. The tone generator provides two 1-kHz outputs, with configurable phase relationship, offering flexibility to create differential signals or test scenarios.

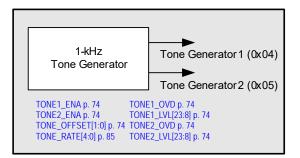


Figure 4-26. Digital-Core Tone Generator

The tone generator outputs can be selected as input to any of the digital mixers or signal-processing functions within the CS42L92 digital core. The hexadecimal numbers in Fig. 4-26 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.



The sample rate for the tone generator is configured using TONE_RATE. See Table 4-26. Note that sample-rate conversion is required when routing the tone generator outputs to any signal chain that is asynchronous or configured for a different sample rate.

The tone generator outputs are enabled by setting the TONE1_ENA and TONE2_ENA bits as described in Table 4-21. The phase relationship is configured using TONE_OFFSET.

The tone generator outputs can also provide a configurable DC signal level, for use as a test signal. The DC output is selected using the TONE n_{OVD} bits, and the DC signal amplitude is configured using the TONE n_{LVL} fields, as described in Table 4-21.

Register Address	Bit	Label	Default	Description
R32 (0x0020)	9:8	TONE_	00	Tone Generator Phase Offset. Sets the phase of Tone Generator 2 relative to Tone
Tone_Generator_1		OFFSET[1:0]		Generator 1
				00 = 0° (in phase)
				01 = 90° ahead
				10 = 180° ahead
				11 = 270° ahead
	5	TONE2_	0	Tone Generator 2 Override
		OVD		0 = Disabled (1-kHz tone output)
				1 = Enabled (DC signal output)
				The DC signal level, when selected, is configured using TONE2_LVL[23:0]
	4	TONE1_	0	Tone Generator 1 Override
		OVD		0 = Disabled (1-kHz tone output)
				1 = Enabled (DC signal output)
				The DC signal level, when selected, is configured using TONE1_LVL[23:0]
	1	TONE2_ENA	0	Tone Generator 2 Enable
				0 = Disabled
				1 = Enabled
	0	TONE1_ENA	0	Tone Generator 1 Enable
				0 = Disabled
				1 = Enabled
R33 (0x0021)	15:0	TONE1_	0x1000	Tone Generator 1 DC output level
Tone_Generator_2		LVL[23:8]		TONE1_LVL[23:8] is coded as 2's complement. Bits [23:20] contain the integer portion; bits [19:0] contain the fractional portion.
				The digital core 0 dBFS level corresponds to 0x10_0000 (+1) or 0xF0_0000 (-1).
R34 (0x0022)	7:0	TONE1_	0x00	Tone Generator 1 DC output level
Tone_Generator_3		LVL[7:0]		TONE1_LVL[23:8] is coded as 2's complement. Bits [23:20] contain the integer portion; bits [19:0] contain the fractional portion.
				The digital core 0 dBFS level corresponds to 0x10_0000 (+1) or 0xF0_0000 (-1).
R35 (0x0023)	15:0	TONE2_	0x1000	
Tone_Generator_4		LVL[23:8]		TONE2_LVL[23:8] is coded as 2's complement. Bits [23:20] contain the integer portion; bits [19:0] contain the fractional portion.
				The digital core 0 dBFS level corresponds to 0x10_0000 (+1) or 0xF0_0000 (-1).
R36 (0x0024)	7:0	TONE2	0x00	Tone Generator 2 DC output level
Tone_Generator_5		LVL[7:0]		TONE2_LVL[23:8] is coded as 2's complement. Bits [23:20] contain the integer portion; bits [19:0] contain the fractional portion.
				The digital core 0 dBFS level corresponds to 0x10_0000 (+1) or 0xF0_0000 (-1).

Table 4-21.	Tone Generator Contr	ol
		.

4.3.10 Noise Generator

The CS42L92 incorporates a white-noise generator that can be routed within the digital core. The main purpose of the noise generator is to provide comfort noise in cases where silence (digital mute) is not desirable.



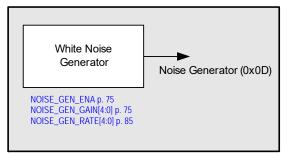


Figure 4-27. Digital-Core Noise Generator

The noise generator can be selected as input to any of the digital mixers or signal-processing functions within the CS42L92 digital core. The hexadecimal number (0x0D) in Fig. 4-27 indicates the corresponding x_SRCn setting for selection of the noise generator as an input to another digital-core function.

The sample rate for the noise generator is configured using NOISE_GEN_RATE. See Table 4-26. Note that sample-rate conversion is required when routing the noise generator output to any signal chain that is asynchronous or configured for a different sample rate.

The noise generator is enabled by setting NOISE_GEN_ENA, described in Table 4-22. The signal level is configured using NOISE_GEN_GAIN.

Register Address	Bit	Label	Default	Description
R160 (0x00A0)	5	NOISE_GEN_	0	Noise Generator Enable
Comfort_Noise_		ENA		0 = Disabled
Generator				1 = Enabled
	4:0	NOISE_GEN_	0x00	Noise generator signal level
		GAIN[4:0]		0x00 = -114 dBFS(6-dB steps) All other codes are reserved
				0x01 = -108 dBFS 0x11 = -6 dBFS
				0x02 = -102 dBFS 0x12 = 0 dBFS

Table 4-22. Noise Generator Control

4.3.11 Haptic Signal Generator

The CS42L92 incorporates a signal generator for use with haptic devices (e.g., mechanical vibration actuators). The haptic signal generator is compatible with both eccentric rotating mass (ERM) and linear resonant actuator (LRA) haptic devices.

The haptic signal generator is highly configurable, and includes the capability to execute a programmable event profile comprising three distinct operating phases.

The resonant frequency of the haptic signal output (for LRA devices) is selectable, providing support for many different actuator components.

The haptic signal generator is a digital signal generator, which is incorporated within the digital core of the CS42L92. In a typical use case the haptic signal may be routed, via one of the digital-core output mixers, to the digital PDM output. An external amplifier can be used to drive the haptic device, as shown in Fig. 4-28.



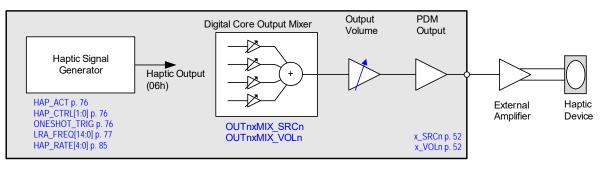


Figure 4-28. Digital-Core Haptic Signal Generator

The hexadecimal number (0x06) in Fig. 4-28 indicates the corresponding x_SRC*n* setting for selection of the haptic signal generator as an input to another digital-core function.

The haptic signal generator is selected as input to one of the digital-core output mixers by setting the x_SRC*n* field of the applicable output mixer to 0x06.

The sample rate for the haptic signal generator is configured using the HAP_RATE field. See Table 4-26. Note that sample-rate conversion is required when routing the haptic signal generator output to any signal chain that is asynchronous or configured for a different sample rate.

The haptic signal generator is configured for an ERM or LRA actuator using the HAP_ACT bit. The required resonant frequency is configured using the LRA_FREQ field. Note that the resonant frequency is only applicable to LRA actuators.

The signal generator can be enabled in continuous mode or configured for one-shot mode using the HAP_CTRL field, as described in Table 4-23. In one-shot mode, the output is triggered by writing to the ONESHOT_TRIG bit.

In one-shot mode, the signal generator profile comprises the distinct phases (1, 2, 3). The duration and intensity of each output phase is programmable.

In continuous mode, the signal intensity is controlled using the PHASE2_INTENSITY field only.

In the case of an ERM actuator (HAP_ACT = 0), the haptic output is a DC signal level, which may be positive or negative, as selected by the x_INTENSITY fields.

For an LRA actuator (HAP_ACT = 1), the haptic output is an AC signal; selecting a negative signal level corresponds to a 180° phase inversion. In some applications, phase inversion may be desirable during the final phase, to halt the physical motion of the haptic device.

Register Address	Bit	Label	Default	Description
R144 (0x0090)	4	ONESHOT_	0	Haptic One-Shot Trigger. Writing 1 starts the one-shot profile (i.e., Phase 1, Phase 2,
Haptics_Control_1		TRIG		Phase 3)
	3:2	HAP_CTRL[1:0]	00	Haptic Signal Generator Control
				00 = Disabled 10 = One-Shot
				01 = Continuous 11 = Reserved
	1	HAP_ACT	0	Haptic Actuator Select
				0 = Eccentric rotating mass (ERM)
				1 = Linear resonant actuator (LRA)

Table 4-23.	Haptic	Signal	Generator	Control
-------------	--------	--------	-----------	---------



Register Address	Bit	Label	Default	Description
R145 (0x0091)	14:0	LRA_	0x7FFF	Haptic Resonant Frequency. Selects the haptic signal frequency (LRA actuator only,
Haptics_Control_2		FREQ[14:0]		HAP_ACT = 1)
				Haptic Frequency (Hz) = System Clock/(2 x (LRA_FREQ+1)), where System Clock = 6.144 MHz or 5.6448 MHz, derived by division from SYSCLK or ASYNCCLK.
				If HAP_RATE < 1000, SYSCLK is the clock source, and the applicable System Clock frequency is determined by SYSCLK.
				If HAP_RATE \geq 1000, ASYNCCLK is the clock source, and the applicable System Clock
				frequency is determined by ASYNCCLK.
				Valid for haptic frequency in the range 100–250 Hz
				For 6.144-MHz System Clock: For 5.6448-MHz System Clock:
				0x77FF = 100 Hz 0x6E3F = 100 Hz
				0x4491 = 175 Hz 0x3EFF = 175 Hz
				0x2FFF = 250 Hz 0x2C18 = 250 Hz
R146 (0x0092)	7:0		0x00	Haptic Output Level (Phase 1). Selects the signal intensity of Phase 1 in one-shot mode.
Haptics_phase_1_		INTENSITY[7:0]		Coded as 2's complement. Range is ± Full Scale (FS).
intensity				For ERM actuator, this selects the DC signal level for the haptic output.
				For LRA actuator, this selects the AC peak amplitude; negative values correspond to a 180° phase shift.
R147 (0x0093)	8:0		0x000	Haptic Output Duration (Phase 1). Selects the duration of Phase 1 in one-shot mode.
Haptics_Control_		DURATION[8:0]		0x000 = 0 ms
phase_1_duration				0x001 = 0.625 ms
				0x002 = 1.25 ms
				(0.625-ms steps)
				0x1FF = 319.375 ms
R148 (0x0094)	7:0		0x00	Haptic Output Level (Phase 2)
Haptics_phase_2_		INTENSITY[7:0]		Selects the signal intensity in Continuous mode or Phase 2 of one-shot mode.
intensity				Coded as 2's complement. Range is ± Full Scale (FS).
				For ERM actuator, this selects the DC signal level for the haptic output.
				For LRA actuator, this selects the AC peak amplitude; Negative values correspond to a 180° phase shift.
R149 (0x0095)	10:0	PHASE2_	0x000	Haptic Output Duration (Phase 2). Selects the duration of Phase 2 in one-shot mode.
Haptics_phase_2_		DURATION[10:0]		0x000 = 0 ms 0x002 = 1.25 ms 0x7FF = 1279.375 ms
duration				0x001 = 0.625 ms (0.625-ms steps)
R150 (0x0096)	7:0	PHASE3_	0x00	Haptic Output Level (Phase 3). Selects the signal intensity of Phase 3 in one-shot mode.
Haptics_phase_3_		INTENSITY[7:0]		Coded as 2's complement.
intensity				Range is ± Full Scale (FS).
				For ERM actuator, this selects the DC signal level for the haptic output.
				For LRA actuator, this selects the AC peak amplitude; Negative values correspond to a 180° phase shift.
R151 (0x0097)	8:0	PHASE3	0x000	Haptic Output Duration (Phase 3). Selects the duration of Phase 3 in one-shot mode.
Haptics_phase_3_	0.0	DURATION[8:0]	5//000	0x000 = 0 ms $0x002 = 1.25 ms$ $0x1FF = 319.375 ms$
duration				0x001 = 0.625 ms (0.625-ms steps)
R152 (0x0098)	0	ONESHOT STS	0	Haptic One-Shot status
Haptics_Status	Ĵ	0.001_0.0	5	0 = One-Shot event not in progress
				1 = One-Shot event in progress

Table 4-23. Haptic Signal Generator Control (Cont.)

4.3.12 PWM Generator

The CS42L92 incorporates two PWM signal generators as shown in Fig. 4-29. The duty cycle of each PWM signal can be modulated by an audio source, or can be set to a fixed value using a control register setting.

A four-input mixer is associated with each PWM generator. The four input sources are selectable in each case, and independent volume control is provided for each path.

PWM signal generators can be output directly on a GPIO pin. See Section 4.14 to configure a GPIO pin for this function.

Note that the PWM signal generators cannot be selected as input to the digital mixers or signal-processing functions within the CS42L92 digital core.



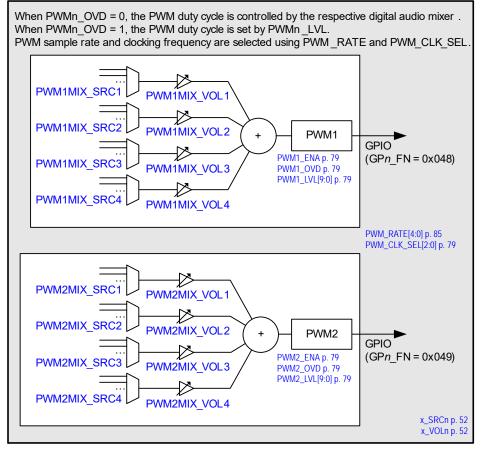


Figure 4-29. Digital-Core PWM Generator

The PWM1 and PWM2 mixer control fields (see Fig. 4-29) are located at register addresses R1600–R1615 (0x0640–0x064F).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the respective mixers. Note that the selected input sources must be configured for the same sample rate as the mixer to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.

The PWM sample rate (cycle time) is configured using PWM_RATE. See Table 4-26. Note that sample-rate conversion is required when linking the PWM generators to any signal chain that is asynchronous or configured for a different sample rate.

The PWM_RATE field must not be changed if any of the associated x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing a new value to PWM_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to PWM_RATE. See Table 4-26 for details.

The PWM generators are enabled by setting PWM1_ENA and PWM2_ENA, respectively, as described in Table 4-24.

Under default conditions ($PWMn_OVD = 0$), the duty cycle of the PWM generators is controlled by an audio signal path; a 4-input mixer is associated with each PWM generator, as shown in Fig. 4-29.

When the PWM*n*_OVD bit is set, the duty cycle of the respective PWM generator is set to a fixed ratio; in this case, the duty cycle ratio is configurable using the PWM*n*_LVL fields.

The PWM generator clock frequency is selected using PWM_CLK_SEL. For best performance, the highest available setting should be used. Note that the PWM generator clock must not be set to a higher frequency than SYSCLK (if PWM_RATE < 1000) or ASYNCCLK (if PWM_RATE \geq 1000).



Register Address	Bit	Label	Default	Description
R48 (0x0030)	10:8	PWM_CLK_	000	PWM Clock Select
PWM_Drive_1		SEL[2:0]		000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				All other codes are reserved.
				The frequencies in brackets apply for 44.1 kHz-related sample rates only.
				PWM_CLK_SEL controls the resolution of the PWM generator; higher settings correspond to higher resolution.
				The PWM Clock must be less than or equal to SYSCLK (if PWM_RATE < 1000) or less than or equal to ASYNCCLK (if PWM_RATE \geq 1000).
	5	PWM2_OVD	0	PWM2 Generator Override
				0 = Disabled (PWM duty cycle is controlled by audio source)
				1 = Enabled (PWM duty cycle is controlled by PWM2_LVL).
	4	PWM1_OVD	0	PWM1 Generator Override
				0 = Disabled (PWM1 duty cycle is controlled by audio source)
				1 = Enabled (PWM1 duty cycle is controlled by PWM1_LVL).
	1	PWM2_ENA	0	PWM2 Generator Enable
				0 = Disabled
				1 = Enabled
	0	PWM1_ENA	0	PWM1 Generator Enable
				0 = Disabled
				1 = Enabled
R49 (0x0031)	9:0	PWM1_LVL[9:0]	0x100	PWM1 Override Level. Sets the PWM1 duty cycle when PWM1_OVD = 1.
PWM_Drive_2				Coded as 2's complement.
				0x000 = 50% duty cycle
				0x200 = 0% duty cycle
R50 (0x0032)	9:0	PWM2_LVL[9:0]	0x100	PWM2 Override Level. Sets the PWM2 duty cycle when PWM2_OVD = 1.
PWM_Drive_3				Coded as 2's complement.
				0x000 = 50% duty cycle
				0x200 = 0% duty cycle

Table 4-24. PWM Generator Control

The CS42L92 automatically checks to confirm that the SYSCLK frequency is high enough to support the digital mixer paths. If an attempt is made to enable a PWM signal mixer path, without sufficient SYSCLK cycles to support it, the attempt fails. Note that any signal paths that are already active are not affected under such circumstances.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

4.3.13 Data Format Conversion

The digital mixing and signal-processing functions on the CS42L92 are designed to route audio data in signed fixed point format. Data format converter (DFC) blocks are incorporated in the digital core, with the capability to convert audio data between signed, unsigned, and floating-point formats. The DFCs enable the flexibility to support many different interface standards on the input and output signal paths. They can also be used to apply dithering to digital audio data.

The digital core provides eight DFC blocks as shown in Fig. 4-30. Each DFC supports one input and one output path only.



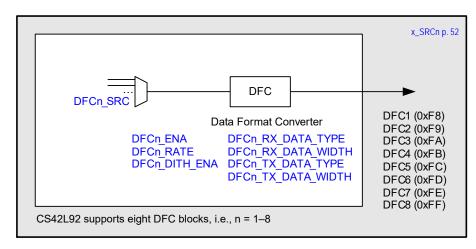


Figure 4-30. Digital-Core DFC Blocks

The DFC1–DFC8 input control fields (see Fig. 4-30) are located at register addresses R3520–R3576 (0x0DC0–0x0DF8).

The full list of digital-mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the respective DFCs. Note that the selected input sources must be configured for the same sample rate as the DFC to which they are connected. Sample-rate conversion functions are available to support flexible interconnectivity; see Section 4.3.15 and Section 4.3.16.

The hexadecimal numbers in Fig. 4-30 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The sample rate for each converter DFC*n* is configured using the respective DFC*n*_RATE field; see Table 4-26. Note that sample-rate conversion is required when routing the DFC paths to any signal chain that is asynchronous or configured for a different sample rate.

The DFC*n*_RATE fields must not be changed if the associated x_SRC*n* field is nonzero. The associated x_SRC*n* field must be cleared before writing a new value to DFC*n*_RATE. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* field and writing to the associated DFC*n*_RATE field. See Table 4-26 for details.

The DFC is enabled by setting DFC*n*_ENA.

The input data format is configured using the DFC*n*_RX_DATA_TYPE and DFC*n*_RX_DATA_WIDTH fields. Valid data types are signed fixed point, unsigned fixed point, and three different floating point configurations. If a fixed point data type is selected, the data width (number of data bits) is selected using DFC*n*_RX_DATA_WIDTH.

The DFC input can be any of the digital core inputs or signal processing blocks. If the input data type is unsigned or floating point format, one of the AIF or SLIMbus RX channels must be selected as the DFC input source—unsigned and floating point data types are not valid selections with any other source within the digital core.

The output data format is configured using the DFCn_TX_DATA_TYPE and DFCn_TX_DATA_WIDTH fields. Valid data types are signed fixed point, unsigned fixed point, and floating point formats. If a fixed point data type is selected, the data width (number of data bits) is selected using DFCn_TX_DATA_WIDTH.

The DFC output can be selected as input to any of the digital mixers or signal-processing functions within the CS42L92 digital core. If the DFC output data type is unsigned fixed point or floating point format, it must be routed directly to the AIF or SLIMbus TX channels using the respective digital-core output mixers—unsigned fixed point or floating point data is not valid as input to any other digital core functions.

Note: If unsigned fixed point or floating point data is routed from a DFC output to an AIF or SLIMbus TX channel, the DFC must be the only enabled signal path in the respective output mixer, and the associated volume selection must be 0 dB.



The DFC can apply dithering to its output data; this is enabled by setting DFC*n*_DITH_ENA. Note that dithering is only valid if the output data is in fixed point format (signed or unsigned). If the output data type is floating point, the DFC*n*_DITH_ENA bit must be cleared.

The dither function can be used to improve the noise characteristics of signals routed in the digital core. Dithering is particularly recommended if truncating audio data (e.g., from 32- to 24-bit format) as it converts the truncation/quantization errors into benign background noise.

The DFCs provide input to the interrupt control circuit and can be used to trigger an interrupt event if saturation (arithmetic error) is detected; see Section 4.15.

The control registers associated with the DFCs are described in Table 4-25.

Register Address	Bit	Label	Default		
R5248 (0x1480)	1	DFC <i>n_</i> DITH_ENA	0	DFC <i>n</i> dither enable (valid for fixed point ou	tput data only)
DFC1_CTRL_W0				0 = Disabled	
R5254 (0x1486) DFC2_CTRL_W0				1 = Enabled	
R5260 (0x148C)	0	DFC <i>n</i> _ENA	0	DFC <i>n</i> enable	
DFC3_CTRL_W0				0 = Disabled	
R5266 (0x1492)				1 = Enabled	
DFC4_CTRL_W0					
R5272 (0x1498)					
DFC5_CTRL_W0 R5278 (0x149E)					
DFC6_CTRL_W0					
R5284 (0x14A4)					
DFC7_CTRL_W0					
R5290 (0x14AA)					
DFC8_CTRL_W0					
R5250 (0x1482)	12:8	DFCn_RX_DATA_ WIDTH[4:0]	0x1F	DFC <i>n</i> input data width (valid for fixed point	
DFC1_RX_W0 R5256 (0x1488)		WIDTH[4.0]			x09 = 10 bits
DFC2 RX W0				0x07 = 8 bits	-20 kits
R5262 (0x148E)	2:0	DFCn RX DATA	000	0x08 = 9 bits 0: DFC <i>n</i> input data type	x1F = 32 bits
DFC3_RX_W0	2.0	TYPE[2:0]	000	000 = Signed, fixed point	
R5268 (0x1494)				001 = Unsigned, fixed point	
DFC4_RX_W0				010 = Single-precision floating point (binary	(32)
R5274 (0x149A) DFC5_RX_W0				100 = Half-precision floating point (binary16	
R5280 (0x14A0)				101 = ARM-alternative half-precision floatin	
DFC6_RX_W0				All other codes are reserved.	
R5286 (0x14A6)					
DFC7_RX_W0					
R5292 (0x14AC)					
DFC8_RX_W0 R5252 (0x1488)	12:8	DFCn TX DATA	0x1F	DFC <i>n</i> output data width (valid for fixed poin	at data types only)
DFC1_TX_W0	12.0	WIDTH[4:0]	0.11		x09 = 10 bits
R5258 (0x148A)		L - J		0x07 = 8 bits	
DFC2_TX_W0					x1F = 32 bits
R5264 (0x1490)	2:0	DFCn_TX_DATA_	000	DFC <i>n</i> output data type	
DFC3_TX_W0		TYPE[2:0]		000 = Signed, fixed point	
R5270 (0x1494) DFC4_TX_W0				001 = Unsigned, fixed point	
R5276 (0x149C)				010 = Single-precision floating point (binary	(32)
DFC5_TX_W0				100 = Half-precision floating point (binary16	
R5282 (0x14A2)				101 = ARM-alternative half-precision floatin	ng point
DFC6_TX_W0				All other codes are reserved.	
R5288 (0x14A8)					
DFC7_TX_W0 R5294 (0x14AE)					
DFC8_TX_W0					
			I		

Table 4-25. Digital-Core DFC Control



The CS42L92 automatically checks to confirm whether the SYSCLK frequency is high enough to support the commanded DFC and digital mixing functions. If an attempt is made to enable DFC signal path, and there are insufficient SYSCLK cycles to support it, the attempt does not succeed. Note that any signal paths that are already active are not affected under such circumstances.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

4.3.14 Sample-Rate Control

The CS42L92 supports multiple signal paths through the digital core. Stereo full-duplex sample-rate conversion is provided to allow digital audio to be routed between interfaces operating at different sample rates and/or referenced to asynchronous clock domains.

Two independent clock domains are supported for the audio signal paths, referenced to SYSCLK and ASYNCCLK respectively, as described in Section 4.16. Every digital signal path must be synchronized either to SYSCLK or to ASYNCCLK.

Up to five different sample rates may be in use at any time on the CS42L92. Three of these sample rates must be synchronized to SYSCLK; the remaining two, where required, must be synchronized to ASYNCCLK.

Sample-rate conversion is required when routing any audio path between digital functions that are asynchronous or configured for different sample rates.

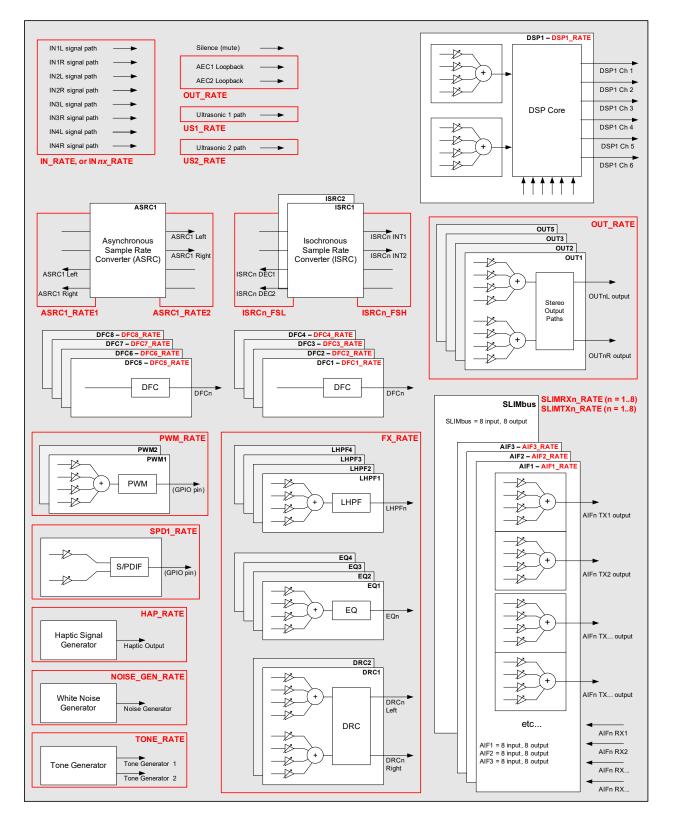
The asynchronous sample-rate converter (ASRC) supports two-way stereo conversion paths between the SYSCLK and ASYNCCLK domains. The ASRC is described in Section 4.3.15.

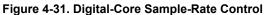
There are two isochronous sample-rate converters (ISRCs). Each ISRC supports two-way, two-channel conversion paths between sample rates on the SYSCLK domain, or between sample rates on the ASYNCCLK domain. The ISRCs are described in Section 4.3.16.

The sample rate of different blocks within the CS42L92 digital core are controlled as shown in Fig. 4-31. The x_RATE fields select the applicable sample rate for each respective group of digital functions.

The x_RATE fields must not be changed if any of the x_SRC*n* fields associated with the respective functions is nonzero. The associated x_SRC*n* fields must be cleared before writing new values to the x_RATE fields. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to the associated x_RATE fields. See Table 4-26 for details.







The input signal paths may be selected as input to the digital mixers or signal-processing functions. The sample rate for the input signal paths can either be set globally (using IN_RATE), or can be configured independently for each input channel (using the respective IN*nx*_RATE fields). The applicable mode depends on IN_RATE_MODE, as described in



Table 4-3.

The ultrasonic demodulator circuits can be selected as input to the digital mixers or signal-processing functions. The sample rate for these signals are configured using US1_RATE and US2_RATE. The selected sample rate must be equal to the output rate of the demodulator function, set by the respective US*n*_FREQ field—see Section 4.2.9.

The output signal paths are derived from the respective output mixers. The sample rate for the output signal paths is configured using OUT_RATE. The sample rate of the AEC loop-back path is also set by OUT_RATE. Clocking for the DACs and output signal path circuits must also be configured using the OUT_CLK_SRC field—see Section 4.11.2.

The AIF*n* RX inputs may be selected as input to the digital mixers or signal-processing functions. The AIF*n* TX outputs are derived from the respective output mixers. The sample rates for digital audio interfaces (AIF1–AIF3) are configured using the AIF*n*_RATE fields (where *n* identifies the applicable AIF 1, 2, or 3) respectively.

The SLIMbus interface supports up to eight input channels and eight output channels. The sample rate of each channel can be configured independently, using SLIMTX*n*_RATE and SLIMRX*n*_RATE.

Note that the SLIMbus interface provides simultaneous support for SYSCLK-referenced and ASYNCCLK-referenced sample rates on different channels. For example, 48-kHz and 44.1-kHz SLIMbus audio paths can be simultaneously supported.

The EQ, DRC, and LHPF functions can be enabled in any signal path within the digital core. The sample rate for these functions is configured using FX_RATE. Note that the EQ, DRC, and LHPF functions must all be configured for the same sample rate.

The DSP functions can be enabled in any signal path within the digital core. The applicable sample rate is configured using the DSP1_RATE field.

The S/PDIF transmitter can be enabled on a GPIO pin. Stereo inputs to this function can be configured from any of the digital-core inputs, mixers, or signal-processing functions. The sample rate of the S/PDIF transmitter is configured using SPD1_RATE.

The tone generator and noise generator can be selected as input to any of the digital mixers or signal-processing functions. The sample rates for these sources are configured using the TONE_RATE and NOISE_GEN_RATE fields, respectively.

The haptic signal generator can be used to control an external vibe actuator. In a typical use case the haptic signal may be routed, via one of the digital-core output mixers, to the digital PDM output (OUT5). The sample rate for the haptic signal generator is configured using HAP_RATE.

The PWM signal generators can be modulated by an audio source, derived from the associated signal mixers. The sample rate (cycle time) for the PWM signal generators is configured using PWM_RATE.

The DFCn blocks can be enabled in signal paths within the digital core. The applicable sample rates are configured using the DFC*n*_RATE fields (where *n* identifies the applicable DFC block).

The sample-rate control registers are described in Table 4-26. Refer to the field descriptions for details of the valid selections in each case—note that the input (ADC) and output (DAC) signal paths must always be associated with the SYSCLK clocking domain; different sample rates may be selected concurrently, but both these rates must be synchronized to SYSCLK.

The control registers associated with the ASRC and ISRCs are described in Table 4-27 and Table 4-28.

Note that 32-bit register addressing is used from R12888 (0x3000) upwards; 16-bit format is used otherwise. The registers noted in Table 4-26 contain a mixture of 16-bit and 32-bit register addresses.



Register Address	Bit	Label	Default	Description
R32 (0x0020)	15:11	TONE_RATE[4:0]	0x00	Tone Generator Sample Rate
Tone_Generator_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
R48 (0x0030)	15:11	PWM_RATE[4:0]	0x00	PWM Frequency (sample rate)
PWM_Drive_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				All PWMnMIX_SRCm fields must be cleared before changing PWM_RATE.
R144 (0x0090)	15:11	HAP_RATE[4:0]	0x00	Haptic Signal Generator Sample Rate
Haptics_Control_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
R160 (0x00A0)	15:11	NOISE_GEN_	0x00	Noise Generator Sample Rate
Comfort_Noise_		RATE[4:0]		0x00 = SAMPLE_RATE_1
Generator				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
R776 (0x0308)	15:11	IN_RATE[4:0]	0x00	Input Signal Paths Sample Rate (only valid if IN_RATE_MODE = 0)
Input_Rate				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				If 384 kHz/768 kHz DMIC rate is selected on any of the input paths (INn_ OSR = 01X), the input paths sample rate is valid up to 48 kHz/96 kHz respectively.

Table 4-26. Digital-Core Sample-Rate Control



Register Address	Bit	Label	Default	Description
R787 (0x0313)	15:11	IN1L_RATE[4:0]	0x00	Input Path <i>n</i> (Left/Right) Sample Rate (only valid if IN_RATE_MODE = 1)
IN1L_Rate_				0x00 = SAMPLE_RATE_1
Control				0x01 = SAMPLE_RATE_2
R791 (0x0317)	15:11	IN1R_RATE[4:0]	0x00	0x02 = SAMPLE_RATE_3
IN1R_Rate_				All other codes are reserved.
Control	45.44		0.00	The selected sample rate is valid in the range 8–192 kHz.
R795 (0x031B)	15:11	IN2L_RATE[4:0]	0x00	If 384 kHz/768 kHz DMIC rate is selected (IN <i>n</i> _OSR = 01X), the IN <i>n</i> L/IN <i>n</i> R sample
IN2L_Rate_ Control				rate is valid up to 48 kHz/96 kHz respectively.
R799 (0x031F)	15.11	IN2R_RATE[4:0]	0x00	
IN2R_Rate_			0.000	
Control				
R803 (0x0323)	15:11	IN3L_RATE[4:0]	0x00	
IN3L_Rate_				
Control				
R807 (0x0327)	15:11	IN3R_RATE[4:0]	0x00	
IN3R_Rate_				
Control	45.44		000	
R811 (0x032B)	15:11	IN4L_RATE[4:0]	0x00	
IN4L_Rate_ Control				
R815 (0x032F)	15:11	IN4R_RATE[4:0]	0x00	
IN4R Rate	10.11		UNU UNU	
Control				
R1032 (0x0408)	15:11	OUT_RATE[4:0]	0x00	Output Signal Paths Sample Rate
Output_Rate_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–384 kHz.
				All OUT <i>nx</i> MIX_SRC <i>m</i> fields must be cleared before changing OUT_RATE.
R1283 (0x0503)	15:11	AIF1_RATE[4:0]	0x00	AIF <i>n</i> Audio Interface Sample Rate
AIF1_Rate_Ctrl				0x00 = SAMPLE_RATE_1
R1347 (0x0543)	15:11	AIF2_RATE[4:0]	0x00	0x01 = SAMPLE_RATE_2
AIF2_Rate_Ctrl			0.00	0x02 = SAMPLE_RATE_3
R1411 (0x0583)	15:11	AIF3_RATE[4:0]	0x00	0x08 = ASYNC_SAMPLE_RATE_1
AIF3_Rate_Ctrl				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–384 kHz.
R1474 (0x05C2)	8:4	SPD1_RATE[4:0]	0x00	All AIF <i>n</i> TXMIX_SRC <i>m</i> fields must be cleared before changing AIF <i>n</i> _RATE.
SPD1_TX_Control	0.4		0,00	S/PDIF Transmitter Sample Rate 0x00 = SAMPLE_RATE_1
				0x00 = SAMPLE_RATE_1 0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x02 = SAMPLE_KATE_S 0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 32–192 kHz.
				All SPDIF1TX n_SRC fields must be cleared before changing SPD1_RATE.
L	I	1	1	

Table 4-26. Digital-Core Sample-Rate Control (Cont.)



Register Address	Bit	Label	Default	Description
R1509 (0x05E5)	15:11		0x00	SLIMbus RX Channel <i>n</i> Sample Rate
SLIMbus_Rates_1		RATE[4:0]		0x00 = SAMPLE_RATE_1
	7:3	SLIMRX1_	0x00	0x01 = SAMPLE_RATE_2
		RATE[4:0]		0x02 = SAMPLE_RATE_3
R1510 (0x05E6)	15:11	SLIMRX4_	0x00	0x08 = ASYNC_SAMPLE_RATE_1
SLIMbus_Rates_2	7.0	RATE[4:0]	0.00	0x09 = ASYNC_SAMPLE_RATE_2
	7:3	SLIMRX3_ RATE[4:0]	0x00	All other codes are reserved.
R1511 (0x05E7)	15:11		0x00	The selected sample rate is valid in the range 8–384 kHz.
SLIMbus Rates 3	10.11	RATE[4:0]	0,00	
CENIDUO_NUICO_O	7:3	SLIMRX5	0x00	
		RATE[4:0]		
R1512 (0x05E8)	14:15	_	0x00	
SLIMbus_Rates_4		RATE[4:0]		
	7:3	SLIMRX7_	0x00	
		RATE[4:0]	0.00	
R1513 (0x05E9)	15:11		0x00	SLIMbus TX Channel <i>n</i> Sample Rate
SLIMbus_Rates_5	7:3	RATE[4:0] SLIMTX1	0x00	0x00 = SAMPLE_RATE_1
	7.5	RATE[4:0]	0,00	0x01 = SAMPLE_RATE_2
R1514 (0x05EA)	15:11		0x00	0x02 = SAMPLE_RATE_3
SLIMbus_Rates_6		RATE[4:0]	0,100	0x08 = ASYNC_SAMPLE_RATE_1
•	7:3	SLIMTX3	0x00	0x09 = ASYNC_SAMPLE_RATE_2
		RATE[4:0]		All other codes are reserved.
R1515 (0x05EB)	15:11		0x00	The selected sample rate is valid in the range 8–384 kHz.
SLIMbus_Rates_7		RATE[4:0]		All SLIMTX <i>n</i> MIX_SRC <i>m</i> fields must be cleared before changing SLIMTX <i>n</i> _RATE.
	7:3	SLIMTX5_	0x00	
	45.44	RATE[4:0]	0.00	
R1516 (0x05EC)	15:11	SLIMTX8_ RATE[4:0]	0x00	
SLIMbus_Rates_8	7:3	SLIMTX7	0x00	
	1.0	RATE[4:0]	0,00	
R3584 (0x0E00)	15:11		0x00	FX Sample Rate (EQ, LHPF, DRC)
FX_Ctrl1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				All EQ <i>n</i> MIX_SRC <i>m</i> , DRC <i>nx</i> MIX_SRC <i>m</i> , and LHPF <i>n</i> MIX_SRC <i>m</i> fields must be
				cleared before changing FX_RATE.
R4225 (0x1081)	15:11	US1_RATE[4:0]	0x00	Ultrasonic Demodulator 1 Sample Rate
US1_Ctrl_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				All other codes are reserved. The selected sample rate must be the same as the output rate set by US1_FREQ (i.e., 8, 16, or 32 kHz).
R4227 (0x1083)	15:11	US2_RATE[4:0]	0x00	Ultrasonic Demodulator 2 Sample Rate
US2_Ctrl_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				All other codes are reserved. The selected sample rate must be the same as the
				output rate set by US2_FREQ (i.e., 8, 16, or 32 kHz).

Table 4-26. Digital-Core Sample-Rate Control (Cont.)



Register Address	Bit	Label	Default	Description
R5248 (0x1480)	6:2	DFC1_RATE[4:0]	0x00	DFC <i>n</i> Sample Rate
DFC1_CTRL_W0				0x00 = SAMPLE_RATE_1
R5254 (0x1486)	6:2	DFC2_RATE[4:0]	0x00	0x01 = SAMPLE_RATE_2
DFC2_CTRL_W0				0x02 = SAMPLE_RATE_3
R5260 (0x148C)	6:2	DFC3_RATE[4:0]	0x00	0x08 = ASYNC SAMPLE RATE 1
DFC3_CTRL_W0				0x09 = ASYNC_SAMPLE_RATE_2
R5266 (0x1492)	6:2	DFC4_RATE[4:0]	0x00	All other codes are reserved.
DFC4_CTRL_W0	0.0	DEOS DATEMAN	0.00	The selected sample rate is valid in the range 8–192 kHz.
R5272 (0x1498)	6:2	DFC5_RATE[4:0]	0x00	The DFC n SRC field must be cleared before changing DFC n RATE.
DFC5_CTRL_W0	0.0		000	
R5278 (0x149E) DFC6_CTRL_W0	6:2	DFC6_RATE[4:0]	0x00	
R5284 (0x14A4)	6:2	DFC7 RATE[4:0]	0x00	
DFC7_CTRL_W0	0.2		0000	
R5290 (0x14AA)	6:2	DFC8 RATE[4:0]	0x00	
DFC8_CTRL_W0	0.2		0,00	
R1048064 (0xF	15:11	DSP1 RATE[4:0]	0x00	DSP1 Sample Rate
FE00)				0x00 = SAMPLE RATE 1
DSP1_Config_1				0x01 = SAMPLE RATE 2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC SAMPLE RATE 1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–384 kHz.
				All DSP1xMIX_SRC <i>m</i> fields must be cleared before changing DSP1_RATE.

Table 4-26. Digital-Core Sample-Rate Control (Cont.)

4.3.15 Asynchronous Sample-Rate Converter (ASRC)

The CS42L92 supports multiple signal paths through the digital core. Two independent clock domains are supported for the audio signal paths, referenced to SYSCLK and ASYNCCLK respectively, as described in Section 4.16. Every digital signal path must be synchronized either to SYSCLK or to ASYNCCLK.

The ASRC provides stereo, two-way signal paths between two sample rates, as shown in Fig. 4-32. Each of the sample rates may be referenced to the SYSCLK or ASYNCCLK domain. The clock domains and sample rates associated with the ASRC signal paths are selected using the ASRC1_RATE1 and ASRC1_RATE2 fields.

- ASRC1_RATE1 selects the clock domain and sample rate of the inputs to the ASRC1 IN1x paths, and the outputs from the ASRC1 IN2x paths.
- ASRC1_RATE2 selects the clock domain and sample rate of the inputs to the ASRC1 IN2x paths, and the outputs from the ASRC1 IN1x paths.

Note that it is possible to select two sample rates for the ASRC that are each referenced to the same clock domain. This provides flexibility to switch between synchronous and asynchronous use cases without changing the signal routing configuration of the affected audio paths.

See Section 4.16 for details of the sample-rate control registers.

The ASRC supports sample rates from 8–192 kHz. The ratio of the applicable SAMPLE_RATE_*n* and ASYNC_SAMPLE_ RATE_*n* fields must not exceed 6.

The ASRC1_RATE1 and ASRC1_RATE2 fields must not be changed if any of the respective x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing new values to ASRC1_RATE1 or ASRC1_RATE2. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to the associated ASRC1_RATE1 or ASRC1_RATE2 fields. See Table 4-27 for details.



The ASRC signal paths are enabled using the ASRC1 INnx ENA bits, as follows:

- The ASRC1 IN1 (left and right) paths convert from the ASRC1_RATE1 sample rate to the ASRC1_RATE2 sample rate. These paths are enabled by setting the ASRC1_IN1L_ENA and ASRC1_IN1R_ENA bits, respectively.
- The ASRC1 IN2 (left and right) paths convert from the ASRC1_RATE2 sample rate to the ASRC1_RATE1 sample rate. These paths are enabled by setting the ASRC1_IN2L_ENA and ASRC1_IN2R_ENA bits, respectively.

Synchronization (lock) between different clock domains is not instantaneous when the clocking or sample rate configurations are updated. The lock status of each ASRC path is an input to the interrupt control circuit and can be used to trigger an interrupt event; see Section 4.15.

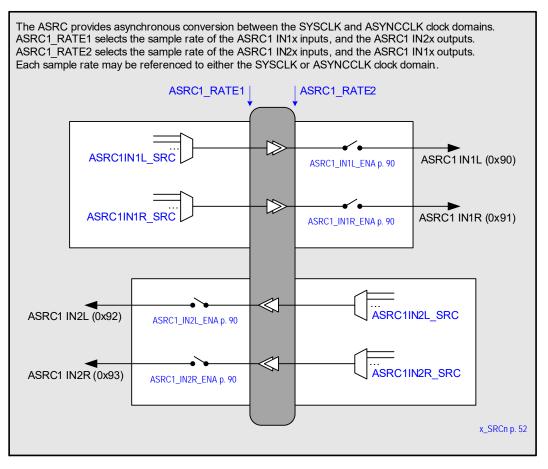
The ASRC lock status of each ASRC path can be output directly on a GPIO pin as an external indication of ASRC lock. See Section 4.14 to configure a GPIO pin for this function.

The CS42L92 performs automatic checks to confirm that the SYSCLK or ASYNCCLK frequency is high enough to support the commanded ASRC and digital mixing functions. If the frequency is too low, an attempt to enable an ASRC signal path fails. Note that active signal paths are not affected under such circumstances.

The status bits in register R3809 indicate the status of each ASRC signal path. If an underclocked error condition occurs, these bits indicate which ASRC signal paths have been enabled.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

The ASRC signal paths and control registers are shown in Fig. 4-32.





The ASRC1 input control fields (see Fig. 4-32) are located at register addresses R2688–R2712 (0x0A80–0x0A98).



The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.

The x_SRC*n* fields select the input sources for the ASRC paths. Note that the selected input sources must be configured for the same sample rate as the ASRC to which they are connected.

The hexadecimal numbers in Fig. 4-32 indicate the corresponding x_SRC*n* setting for selection of that signal as an input to another digital-core function.

The fields associated with the ASRC are described in Table 4-27.

Register Address	Bit	Label	Default	Description
R3808 (0x0EE0)	3	ASRC1_IN2L_	0	ASRC1 IN2 (left) enable
ASRC1_ENABLE		ENA		(Left channel from ASRC1_RATE2 sample rate to ASRC1_RATE1 sample rate)
				0 = Disabled
				1 = Enabled
	2	ASRC1_IN2R_	0	ASRC1 IN2 (right) enable
		ENA		(Right channel from ASRC1_RATE2 sample rate to ASRC1_RATE1 sample rate)
				0 = Disabled
				1 = Enabled
	1	ASRC1_IN1L_	0	ASRC1 IN1 (left) enable
		ENA		(Left channel from ASRC1_RATE1 sample rate to ASRC1_RATE2 sample rate)
				0 = Disabled
				1 = Enabled
	0	ASRC1_IN1R_	0	ASRC1 IN1 (right) enable
		ENA		(Right channel from ASRC1_RATE1 sample rate to ASRC1_RATE2 sample rate)
				0 = Disabled
				1 = Enabled

Table 4-27. Digital-Core ASRC Control



Register Address	Bit	Label	Default	Description
R3809 (0x0EE1)	3	ASRC1_IN2L_	0	ASRC1 IN2 (left) enable status
ASRC1_STATUS		ENA_STS		(Left channel from ASRC1_RATE2 sample rate to ASRC1_RATE1 sample rate)
				0 = Disabled
				1 = Enabled
	2	ASRC1_IN2R_	0	ASRC1 IN2 (right) enable status
		ENA_STS		(Right channel from ASRC1_RATE2 sample rate to ASRC1_RATE1 sample rate)
				0 = Disabled
				1 = Enabled
	1	ASRC1_IN1L_	0	ASRC1 IN1 (left) enable status
		ENA_STS		(Left channel from ASRC1_RATE1 sample rate to ASRC1_RATE2 sample rate)
				0 = Disabled
				1 = Enabled
	0	ASRC1_IN1R_	0	ASRC1 IN1 (right) enable status
		ENA_STS		(Right channel from ASRC1_RATE1 sample rate to ASRC1_RATE2 sample rate)
				0 = Disabled
				1 = Enabled
R3810 (0x0EE2)			0x00	ASRC1 Sample Rate select for ASRC1 IN1x inputs and ASRC1 IN2x outputs
ASRC1_RATE1		RATE1[4:0]		0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				All ASRC1_IN1x_SRC fields must be cleared before changing ASRC1_RATE1.
R3811 (0x0EE3)	15:11	ASRC1_	0x08	ASRC1 Sample Rate select for ASRC1 IN2x inputs and ASRC1 IN1x outputs
ASRC1_RATE2		RATE2[4:0]		0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				All ASRC1_IN1x_SRC fields must be cleared before changing ASRC1_RATE1.

Table 4-27. Digital-Core ASRC Control (Cont.)

4.3.16 Isochronous Sample-Rate Converter (ISRC)

The CS42L92 supports multiple signal paths through the digital core. The ISRCs provide sample-rate conversion between synchronized sample rates on the SYSCLK clock domain, or between synchronized sample rates on the ASYNCCLK clock domain.

There are two ISRCs on the CS42L92. Each ISRC provides two stereo signal paths between two different sample rates, as shown in Fig. 4-33.

The sample rates associated with each ISRC can be set independently. Note that the two sample rates associated with any single ISRC must both be referenced to the same clock domain (SYSCLK or ASYNCCLK).

- When an ISRC is used on the SYSCLK domain, the associated sample rates may be selected from SAMPLE_ RATE_1, SAMPLE_RATE_2, or SAMPLE_RATE_3.
- When an ISRC is used on the ASYNCCLK domain, the associated sample rates are ASYNC_SAMPLE_RATE_1 and ASYNC_SAMPLE_RATE_2.

See Section 4.16 for details of the sample-rate control registers.

Each ISRC converts between a sample rate selected by ISRC*n*_FSL and a sample rate selected by ISRC*n*_FSH, (where *n* identifies the applicable ISRC 1 or 2). The higher of the two sample rates must be selected by ISRC*n*_FSH in each case.



The ISRCs support sample rates in the range 8–384 kHz. For each ISRC, the ratio of the applicable SAMPLE_RATE_*n* or ASYNC_SAMPLE_RATE_*n* fields must not exceed 24. The sample-rate conversion ratio must be an integer (1–24) or equal to 1.5.

The ISRC*n*_FSL and ISRC*n*_FSH fields must not be changed if any of the respective x_SRC*n* fields is nonzero. The associated x_SRC*n* fields must be cleared before writing new values to ISRC*n*_FSL or ISRC*n*_FSH. A minimum delay of 125 μ s must be allowed between clearing the x_SRC*n* fields and writing to the associated ISRC*n*_FSL or ISRC*n*_FSH fields. See Table 4-28 for details.

The ISRC signal paths are enabled using the ISRC*n*_INT*m*_ENA and ISRC*n*_DEC*m*_ENA bits, as follows:

- The ISRC*n* interpolation paths (increasing sample rate) are enabled by setting the ISRC*n*_INT*m*_ENA bits, (where *m* identifies the applicable channel).
- The ISRC*n* decimation paths (decreasing sample rate) are enabled by setting the ISRC*n*_DEC*m*_ENA bits.

The CS42L92 performs automatic checks to confirm that the SYSCLK or ASYNCCLK frequency is high enough to support the commanded ISRC and digital mixing functions. If the frequency is too low, an attempt to enable an ISRC signal path fails. Note that active signal paths are not affected under such circumstances.

The status bits in registers R1600–R3576 indicate the status of each digital mixer. If an underclocked error condition occurs, these bits indicate which mixers have been enabled.

The ISRC signal paths and control registers are shown in Fig. 4-33.

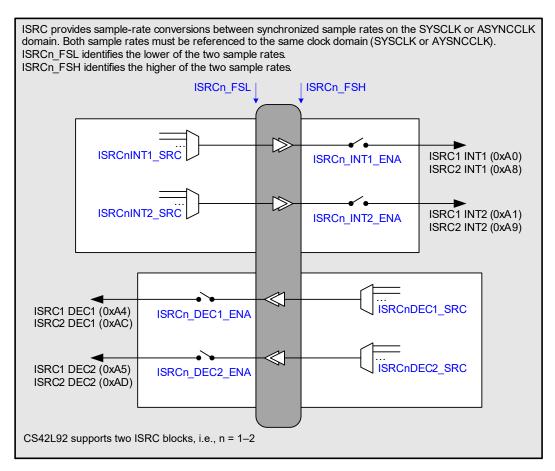


Figure 4-33. Isochronous Sample-Rate Converters (ISRCs)

The ISRC input control fields (see Fig. 4-33) are located at register addresses R2816–R2920 (0x0B00–0x0B68).

The full list of digital mixer control registers (R1600–R3576) is provided in Section 6. Generic register field definitions are provided in Table 4-11.



The x_SRC fields select the input sources for the respective ISRC processing blocks. Note that the selected input sources must be configured for the same sample rate as the ISRC to which they are connected.

The hexadecimal numbers in Fig. 4-33 indicate the corresponding x_SRC setting for selection of that signal as an input to another digital-core function.

The fields associated with the ISRCs are described in Table 4-28.

Register Address	Bit	Label	Default	Description
R3824 (0x0EF0)	15:11	ISRC1_FSH[4:0]	0x00	ISRC1 High Sample Rate (Sets the higher of the ISRC1 sample rates)
ISRC1_CTRL_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				The ISRC1_FSH and ISRC1_FSL fields must both select sample rates referenced to the same clock domain (SYSCLK or ASYNCCLK).
				All ISRC1_DEC <i>n</i> _SRC fields must be cleared before changing ISRC1_FSH.
R3825 (0x0EF1)	15:11	ISRC1_FSL[4:0]	0x00	ISRC1 Low Sample Rate (Sets the lower of the ISRC1 sample rates)
ISRC1_CTRL_2				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE RATE 2
				0x02 = SAMPLE RATE 3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				The ISRC1 FSH and ISRC1 FSL fields must both select sample rates referenced to
				the same clock domain (SYSCLK or ASYNCCLK).
				All ISRC1_INT <i>n</i> _SRC fields must be cleared before changing ISRC1_FSL.
R3826 (0x0EF2)	15	ISRC1_INT1_ENA	0	ISRC1 INT1 Enable (Interpolation Channel 1 path from ISRC1_FSL rate to ISRC1_
ISRC1_CTRL_3				FSH rate)
				0 = Disabled
				1 = Enabled
	14	ISRC1_INT2_ENA	0	ISRC1 INT2 Enable (Interpolation Channel 2 path from ISRC1_FSL rate to ISRC1_ FSH rate)
				0 = Disabled
				1 = Enabled
	9	ISRC1_DEC1_ ENA	0	ISRC1 DEC1 Enable (Decimation Channel 1 path from ISRC1_FSH rate to ISRC1_ FSL rate)
				0 = Disabled
				1 = Enabled
	8	ISRC1_DEC2_ ENA	0	ISRC1 DEC2 Enable (Decimation Channel 2 path from ISRC1_FSH rate to ISRC1_ FSL rate)
				0 = Disabled
				1 = Enabled
R3827 (0x0EF3)	15:11	ISRC2_FSH[4:0]	0x00	ISRC2 High Sample Rate (Sets the higher of the ISRC2 sample rates)
ISRC2_CTRL_1				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				The ISRC2_FSH and ISRC2_FSL fields must both select sample rates referenced to the same clock domain (SYSCLK or ASYNCCLK).
				All ISRC2_DEC <i>n</i> _SRC fields must be cleared before changing ISRC2_FSH.

Table 4-28. Digital-Core ISRC Control



Register Address	Bit	Label	Default	Description
R3828 (0x0EF4)	15:11	ISRC2_FSL[4:0]	0x00	ISRC2 Low Sample Rate (Sets the lower of the ISRC2 sample rates)
ISRC2_CTRL_2				0x00 = SAMPLE_RATE_1
				0x01 = SAMPLE_RATE_2
				0x02 = SAMPLE_RATE_3
				0x08 = ASYNC_SAMPLE_RATE_1
				0x09 = ASYNC_SAMPLE_RATE_2
				All other codes are reserved.
				The selected sample rate is valid in the range 8–192 kHz.
				The ISRC2_FSH and ISRC2_FSL fields must both select sample rates referenced to the same clock domain (SYSCLK or ASYNCCLK).
				All ISRC2_INT <i>n</i> _SRC fields must be cleared before changing ISRC2_FSL.
R3829 (0x0EF5)	15	ISRC2_INT1_ENA	0	ISRC2 INT1 Enable (Interpolation Channel 1 path from ISRC2_FSL rate to ISRC2_
ISRC2_CTRL_3				FSH rate)
				0 = Disabled
				1 = Enabled
	14	ISRC2_INT2_ENA	0	ISRC2 INT2 Enable (Interpolation Channel 2 path from ISRC2_FSL rate to ISRC2_ FSH rate)
				0 = Disabled
				1 = Enabled
	9	ISRC2_DEC1_ ENA	0	ISRC2 DEC1 Enable (Decimation Channel 1 path from ISRC2_FSH rate to ISRC2_ FSL rate)
				0 = Disabled
				1 = Enabled
	8	ISRC2_DEC2_	0	ISRC2 DEC2 Enable (Decimation Channel 2 path from ISRC2_FSH rate to ISRC2_
		ENA		FSL rate)
				0 = Disabled
				1 = Enabled

Table 4-28. Digital-Core ISRC Control (Cont.)

4.4 DSP Firmware Control

The CS42L92 digital core incorporates one programmable DSP processing block, capable of running a range of application-specific algorithms. Different firmware configurations can be loaded onto the DSP, enabling the CS42L92 to be customized for specific application requirements. Full read/write access to the device register map is supported from the DSP core.

The DSP can be clocked at up to 75 MHz, corresponding to 75 MIPS. A software programming guide can be provided to assist users in developing their own software algorithms—please contact your Cirrus Logic representative for further information.

To use the programmable DSP, the required firmware configuration must first be loaded onto the device by writing the appropriate files to the CS42L92 register map. The firmware configuration comprises program, data, and coefficient content.

Details of the DSP firmware memory registers are provided in Section 4.4.1. Note that the WISCE evaluation board control software provides support for easy loading of program, data, and coefficient content onto the CS42L92. Please contact your Cirrus Logic representative for more details of the WISCE evaluation board control software.

After loading the DSP firmware, the DSP functions must be enabled using the associated control fields.

The audio signal paths to and from the DSP processing block are configured as described in Section 4.3. Note that the DSP firmware must be loaded and enabled before audio signal paths can be enabled.

4.4.1 DSP Firmware Memory and Register Mapping

The DSP firmware memory is programmed by writing to the registers referenced in Table 4-29. Note that clocking is not required for access to the firmware registers by the host processor.



The CS42L92 program, data, and coefficient register memory space is described in Table 4-29. The full register map listing is provided in Section 6.

The program firmware parameters are formatted as 40-bit words. For this reason, 3 x 32-bit register addresses are required for every 2 x 40-bit words.

DSP Number	Description	Register Address	Number of Registers	DSP Memory Size
DSP1	Program memory	0x08_0000-0x08_2FFE	6144	4k x 40-bit words
	X-Data memory	0x0A_0000-0x0A_1FFE	4096	4k x 24-bit words
	Y-Data memory	0x0C_0000-0x0C_1FFE	4096	4k x 24-bit words
	Coefficient memory	0x0E_0000-0x0E_1FFE	4096	4k x 24-bit words

Table 4-29. DSP Program, Data, and Coefficient Registers

The X-memory on the DSP supports read/write access to all register fields throughout the device, including the codec control registers, and the other firmware-memory regions of DSP core itself. Access to the register address space is supported using a number of register windows within the X-memory on the DSP.

Note that the register window space is additional to the X-data memory size described in Table 4-29.

Addresses 0xC000 to 0xDFFF in X-memory map directly to addresses 0x0000 to 0x1FFF in the device register space. This fixed register window contains primarily the codec control registers; it also includes the virtual DSP control registers (described in Section 4.4.7). Each X-memory address within this window maps onto one 16-bit register in the codec memory space.

Four movable register windows are also provided, starting at X-memory addresses 0xF000, 0xF400, 0xF800, and 0xFC00 respectively. Each window represents 1024 addresses in the X-memory space. The start address, within the corresponding device register space, for each window is configured using DSP1_EXT_[A/B/C/D]_PAGE (where A defines the first window, B defines the second window, etc.).

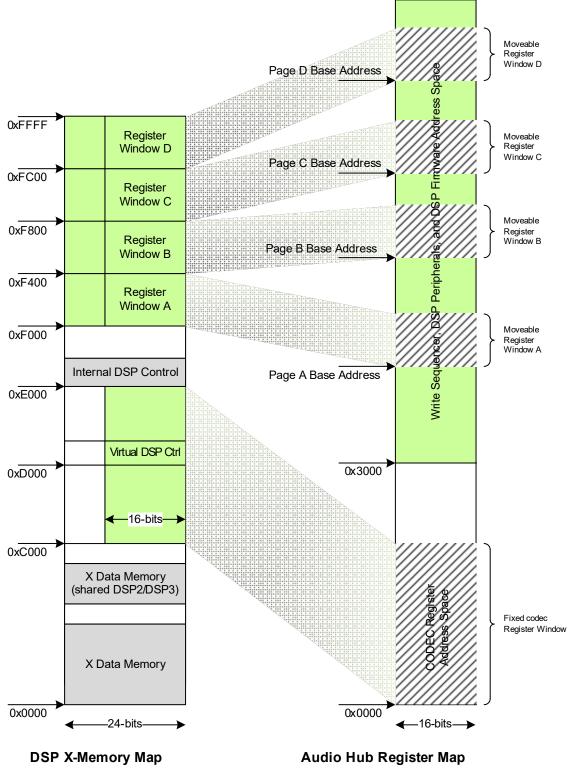
Two mapping modes are supported and are selected using the DSP1_EXT_[A/B/C/D]_PSIZE16 bits for the respective window. In 16-Bit Mode, each address within the window maps onto one 16-bit register in the device memory space; the window equates to 1024 x 16-bit registers. In 32-Bit Mode, each address within the window maps onto two 16-bit registers in the device memory space; the window equates to 1024 x 32-bit registers.

Note that the X-memory is only 24-bits wide; as a result, the upper 8 bits of the odd-numbered register addresses are not mapped, and cannot be accessed, in 32-Bit Mode.

The DSP1_EXT_[A/B/C/D]_PAGE fields are defined with an LSB = 512. Accordingly, the base address of each window must be aligned with 512-word boundaries. Note that the base addresses are entirely independent of each other; for example, overlapping windows are permissible if required, and there is no requirement for the A/B/C/D windows to be at incremental locations.

The register map window functions are shown in Fig. 4-34. Further information on the definition and usage of the DSP firmware memories is provided in the software programming guide; contact your Cirrus Logic representative if required.







Note that the full CS42L92 register space is shown here as 16-bit width. (SPI/I²C register access uses 32-bit data width at 0x3000 and above.) However, the window base address fields (DSP1_EXT_[A/B/C/D]_PAGE) are referenced to 16-bit width, and 16-bit register mapping is shown. Hence, the device register map is shown here entirely as 16-bit width for ease of explanation.

The control registers associated with the register map window functions are described in Table 4-30.



Register Address	Bit	Label	Default	Description
R1048148 (0xF_FE54)	31	DSP1_EXT_A_PSIZE16	0	Register Window A page width select
DSP1_Ext_window_A				0 = 32-bit
				1 = 16-bit
				Note that, in 32-Bit Mode, only the lower 24 bits can be accessed.
	15:0	DSP1_EXT_A_PAGE[15:0]	0x0000	Sets the Base Address of Register Window A in X-memory.
				Coded as LSB = 512 (0x200)
R1048150 (0xF_FE56)	31	DSP1_EXT_B_PSIZE16	0	Register Window B page width select
DSP1_Ext_window_B				0 = 32-bit
				1 = 16-bit
				Note that, in 32-Bit Mode, only the lower 24 bits can be accessed.
	15:0	DSP1_EXT_B_PAGE[15:0]	0x0000	Sets the Base Address of Register Window B in X-memory.
				Coded as LSB = 512 (0x200)
R1048152 (0xF_FE58)	31	DSP1_EXT_C_PSIZE16	0	Register Window C page width select
DSP1_Ext_window_C				0 = 32-bit
				1 = 16-bit
				Note that, in 32-Bit Mode, only the lower 24 bits can be accessed.
	15:0	DSP1_EXT_C_PAGE[15:0]	0x0000	Sets the Base Address of Register Window C in X-memory.
				Coded as LSB = 512 (0x200)
R1048154 (0xF_FE5A)	31	DSP1_EXT_D_PSIZE16	0	Register Window D page width select
DSP1_Ext_window_D				0 = 32-bit
				1 = 16-bit
				Note that, in 32-Bit Mode, only the lower 24 bits can be accessed.
	15:0	DSP1_EXT_D_PAGE[15:0]	0x0000	Sets the Base Address of Register Window D in X-memory.
				Coded as LSB = 512 (0x200)

Table 4-30. X-Data Memory and Clocking Control

4.4.2 DSP Memory Locking

The DSP core has the capability for read/write access to all register fields throughout the device, including the codec control registers, DSP peripheral control registers, and the virtual DSP control registers. Access to these registers is supported via the DSP's X-memory (using the register windows), as described in Section 4.4.1.

The CS42L92 provides a register-locking feature that blocks DSP register-write attempts to invalid register regions, preventing the firmware from making unintentional changes to register and memory contents. An interrupt event and associated debug information are generated if any write-access attempt is blocked; this can be used to assist software development and debug.

The register map and DSP firmware memories are partitioned into four regions; each region can be locked independently. This allows full flexibility to lock different register/memory regions according to the applicable DSP firmware configuration.

The DSP has direct access to its own X-, Y-, Z-, and P- memories; this is always enabled and cannot be locked. Access to the codec registers, DSP peripheral registers, and the virtual DSP registers is effected using the X-memory register windows (fixed codec window, and four configurable windows)—write access to these locations is governed by the register-locking configuration settings.

The virtual DSP registers occupy addresses within the codec register space; these registers represent one of the lockable regions within the register map—two independent locks are provided for the codec and virtual DSP registers.

Note: A DSP register window can be mapped onto the X-, Y-, Z-, or P- memory region of the DSP. In this event, write access via that window is governed by the register locks, potentially blocking the DSP from accessing its own memory. This is not the intended use of the register lock, however.

The lockable register/memory regions are defined in Table 4-31.



Table 4-31. DSP Memory Locking Regions

Region	Description	Register Address	Notes
Region 0	Virtual DSP registers	0x00_1000-0x00_2FFF	Excludes memory lock and watchdog reset registers
Region 1	Codec registers	0x00_0000-0x03_FFFE	Excludes virtual DSP registers
Region 2	DSP peripheral control registers	0x04_0000-0x07_FFFE	—
Region 3	DSP1 memory	0x08_0000-0x09_FFFE	—

The register locks are controlled using the DSP1_CTRL_REGION*m*_LOCK fields (where *m* identifies the register/memory region). The associated lock determines whether the DSP core is granted write access to region *m*. To change the lock status, two writes must be made to the respective register field:

- Writing 0x5555, followed by 0xAAAA, sets the respective lock
- Writing 0xCCCC, followed by 0x3333, clears the respective lock

The status of each lock can be read from the DSP1_CTRL_REGION*m*_LOCK_STS bits.

Write access to the DSP1_CTRL_REGION*m*_LOCK fields is always possible. This means that the DSP core always has write access for configuring the memory-access locks.

The DSP memory locking function is an input to the interrupt control circuit and can be used to trigger an interrupt event if an invalid register write is attempted—see Section 4.4.5. Additional status and control fields are provided for debug purposes, as described in Section 4.4.6.

The control registers associated with the DSP memory locking functions are described in Table 4-32.

Register Address	Bit	Label	Default	Description
R1048164 (0xF_FE64)	3	DSP1_CTRL_REGION3_LOCK_STS	0	DSP1 memory region <i>m</i> lock status
DSP1_Region_lock_sts_0	2	DSP1_CTRL_REGION2_LOCK_STS	0	0 = Unlocked
	1	DSP1_CTRL_REGION1_LOCK_STS	0	1 = Locked (write access is blocked)
	0	DSP1_CTRL_REGION0_LOCK_STS	0	
R1048166 (0xF_FE66)	31:16	DSP1_CTRL_REGION1_LOCK[15:0]	See	DSP1 memory region <i>m</i> lock.
DSP1_Region_lock_1			Footnote 1	Write 0x5555, then 0xAAAA, to set the lock.
DSP1_Region_lock_0	15:0	DSP1_CTRL_REGION0_LOCK[15:0]	See	Write 0xCCCC, then 0x3333, to clear the lock.
			Footnote 1	
R1048168 (0xF_FE68)	31:16	DSP1_CTRL_REGION3_LOCK[15:0]	See	
DSP1 Region lock 3			Footnote 1	
DSP1_Region_lock_2	15:0	DSP1_CTRL_REGION2_LOCK[15:0]	See	
			Footnote 1	

Table 4-32. DSP Memory Locking Control

1. Default is not applicable to these write-only fields

4.4.3 DSP Firmware Control

The configuration and control of the DSP firmware is described in the following subsections.

4.4.3.1 DSP Memory

The DSP memory (program, X-data, Y-data, and coefficient) is enabled by setting DSP1_MEM_ENA. This memory must be enabled (DSP1_MEM_ENA = 1) for read/write access, code execution, and DMA functions. The DSP memory is disabled, and the contents lost, whenever the DSP1_MEM_ENA bit is cleared.

The DSP1_RAM_RDY status bit indicates whether the DSP memory is ready for read/write access. The DSP memory should not be accessed until this bit has been set.

The DSP1_MEM_ENA bit is not affected by software reset; it remains in its previous state under software reset conditions. Accordingly, the DSP memory contents are maintained through software reset, provided DCVDD is held above its reset threshold.

The DSP firmware memory is always cleared under power-on reset, hardware reset, and Sleep Mode conditions. See Section 5.2 for a summary of the CS42L92 reset behavior.



4.4.3.2 DSP Clocking

Clocking is required for the DSP processing block, when executing software or when supporting DMA functions. (Note that clocking is not required for access to the firmware registers by the host processor.) Clocking within the DSP is enabled and disabled automatically, as required by the DSP core and DMA channel status.

The clock source for each DSP is derived from DSPCLK. See Section 4.16 for details of how to configure DSPCLK.

The clock frequency for the DSP is selected using DSP1_CLK_FREQ_SEL. The DSP clock frequency must be less than or equal to the DSPCLK frequency.

The DSP1_CLK_FREQ_STS field indicates the clock frequency for the DSP core. This can be used to confirm the clock frequency, in cases where code execution has a minimum clock frequency requirement. The DSP1_CLK_FREQ_STS field is only valid when the core is running code; typical usage of this field would be for the DSP core itself to read the clock status and to take action as applicable, in particular, if the available clock does not meet the application requirements.

Note that, depending on the DSPCLK frequency and the available clock dividers, the DSP1 clock frequency may differ from the selected clock. In most cases, the DSP1 clock frequency equals or exceeds the requested frequency. A lower frequency is implemented if limited by either the DSPCLK frequency or the maximum DSP1 clocking frequency.

The DSPCLK configuration provides input to the interrupt control circuit and can be used to trigger an interrupt event when the DSP1 clock frequency is less than the requested frequency; see Section 4.15.

4.4.3.3 DSP Code Execution

After the DSP firmware has been loaded, and the clocks configured, the DSP block is enabled by setting DSP1_CORE_ENA. When the DSP is configured and enabled, the firmware execution can be started by writing 1 to DSP1_START.

Alternative methods to trigger the firmware execution can also be configured using the DSP1_START_IN_SEL field.

Using the DSP1_START_IN_SEL field, the DSP firmware execution can be linked to the respective DMA function, the IRQ2 status, or to the FIFO status in the event logger:

- DMA function: firmware execution commences when all enabled DSP input (WDMA) channel buffers have been filled, and all enabled DSP output (RDMA) channel buffers have been emptied
- IRQ2: firmware execution commences when one or more of the unmasked IRQ2 events has occurred
- Event logger status: firmware execution commences when the FIFO not-empty status is asserted within the event logger

To enable firmware execution on the DSP block, the DSP1_CORE_ENA bit must be set. Note that the usage of the DSP1_ START bit may vary depending on the particular firmware that is being executed: in some applications (e.g., when an alternative trigger is selected using DSP1_START_IN_SEL), writing to the DSP1_START bit is not required.

4.4.3.4 DSP Watchdog Timer

A watchdog timer is provided for the DSP, which can be used to detect software lock-ups, and other conditions that require corrective action in order to resume the intended DSP behavior.

The DSP1 watchdog is enabled using DSP1_WDT_ENA. The timeout period is configured using DSP1_WDT_MAX_COUNT.

In normal operation, the watchdog should be reset regularly—this action is used to confirm that the DSP code is running correctly. The watchdog is reset by writing 0x5555, followed by 0xAAAA, to the DSP1_WDT_RESET field.

The watchdog status bit, DSP1_WDT_TIMEOUT_STS, is set if the timeout period elapses before the watchdog is reset; this event typically signals that a lock-up or other error condition has occurred.

The DSP watchdog is an input to the interrupt control circuit and can be used to trigger an interrupt event if the timeout period elapses—see Section 4.4.5.

Note that write access to the DSP1_WDT_RESET field is not affected by the register locking mechanism (see Section 4.4.2). This means that the DSP core always has write access to reset the watchdog.



4.4.3.5 DSP Control Registers

The DSP memory, clocking, code-execution, and watchdog control registers are described in Table 4-33.

The audio signal paths connecting to/from the DSP processing block are configured as described in Section 4.3. Note that the DSP firmware must be loaded and enabled before audio signal paths can be enabled.

Register Address	Bit	Label	Default	Description
R1048064 (0xF_	4	DSP1_MEM_ENA	0	DSP1 memory control
FE00)				0 = Disabled
DSP1_Config_1				1 = Enabled
				The DSP1 memory contents are lost when DSP1_MEM_ENA =0. Note that this bit is not affected by software reset; it remains in its previous condition.
	1	DSP1_CORE_ENA	0	DSP1 enable. Controls the DSP1 firmware execution
				0 = Disabled
				1 = Enabled
	0	DSP1_START	—	DSP1 start
				Write 1 to start DSP1 firmware execution
R1048066 (0xF_	15:0	DSP1_CLK_FREQ_	0x0000	DSP1 clock frequency select
FE02)		SEL[15:0]		Coded as LSB = 1/64 MHz, Valid from 5.6 to 75 MHz.
DSP1_Config_2				The DSP1 clock must be less than or equal to the DSPCLK frequency. The DSP1 clock is generated by division of DSPCLK, and may differ from the selected frequency. The DSP1 clock frequency can be read from DSP1_CLK_FREQ_STS.
R1048070 (0xF_	0	DSP1_CLK_AVAIL	0	DSP1 clock availability (read only)
FE06)				0 = No Clock
DSP1_Status_2				1 = Clock Available
				This bit exists for legacy software support only; it is not recommended for future designs—it may be unreliable on the latest device architectures.
R1048072 (0xF_ FE08)	15:0	DSP1_CLK_FREQ_ STS[15:0]	0x0000	DSP1 clock frequency (read only). Valid only when the respective DSP core is enabled.
DSP1_Status_3				Coded as LSB = 1/64 MHz.
R1048074 (0xF_	4:1	DSP1_WDT_MAX_	0x0	DSP1 watchdog timeout value.
FE0A)		COUNT[3:0]		0x0 = 2 ms 0x5 = 64 ms 0xA = 2 s
DSP1_Watchdog_1				0x1 = 4 ms 0x6 = 128 ms 0xB = 4 s
				0x2 = 8 ms 0x7 = 256 ms 0xC = 8 s
				0x3 = 16 ms 0x8 = 512 ms 0xD–0xF = reserved
				0x4 = 32 ms 0x9 = 1 s
	0	DSP1_WDT_ENA	0	DSP1 watchdog enable
				0 = Disabled
				1 = Enabled
R1048120 (0xF_	4:0	DSP1_START_IN_	0x00	DSP1 firmware execution control. Selects the trigger for DSP1 firmware
FE38)		SEL[4:0]		execution. 0x00 = DMA 0x0B = IRQ2 0x10 = Event Logger 1
DSP1_External_Start				0x00 = DMA 0x0B = IRQ2 0x10 = Event Logger 1 All other codes are reserved.
				Note that the DSP1_START bit also starts the DSP1 firmware execution, regardless of this field setting.
R1048158 (0xF_ FE5E)	15:0	DSP1_WDT_ RESET[15:0]	0x0000	DSP1 watchdog reset.
,				Write 0x5555, followed by 0xAAAA, to reset the watchdog.
DSP1_Watchdog_2	10			DSD1 watchdog timogut atotus
R1048186 (0xF_ FE7A)	13	DSP1_WDT_ TIMEOUT_STS	0	DSP1 watchdog timeout status
DSP1_Region_lock_ ctrl_0				This bit, when set, indicates that the watchdog timeout has occurred. This bit is latched when set; it is cleared when the watchdog is disabled or reset.

Table 4-33.	DSP	Control	Registers
-------------	-----	---------	-----------

4.4.4 DSP Direct Memory Access (DMA) Control

The DSP provides a multichannel DMA function; this is configured using the registers described in Table 4-34.



There are eight WDMA (DSP input) and six RDMA (DSP output) channels; these are enabled using the DSP1_WDMA_ CHANNEL_ENABLE and DSP1_RDMA_CHANNEL_ENABLE fields. The status of each WDMA channel is indicated in DSP1_WDMA_ACTIVE_CHANNELS.

The DMA can access the X-data memory or Y-data memory associated with the DSP block. The applicable memory is selected using bit [15] of the respective x_START_ADDRESS field for each DMA channel.

The start address of each DMA channel is configured as described in Table 4-34. Note that the required address is defined relative to the base address of the selected (X-data or Y-data) memory.

The buffer length of the DMA channels is configured using the DSP1_DMA_BUFFER_LENGTH field. The selected buffer length applies to all enabled DMA channels.

Note that the start-address fields and buffer-length fields are defined in 24-bit DSP data word units. This means that the LSB of these fields represents one 24-bit DSP memory word. This differs from the CS42L92 register map layout described in Table 4-29.

The parameters of a DMA channel (i.e., start address or offset address) must not be changed while the respective DMA is enabled. All of the DMA channels must be disabled before changing the DMA buffer length.

Each DMA channel uses a twin buffer mechanism to support uninterrupted data flow through the DSP. The buffers are called *ping* and *pong*, and are of configurable size, as noted above. Data is transferred to/from each of the buffers in turn.

When the ping input data buffer is full, the DSP1_PING_FULL bit is set, and a DSP start signal is generated. The start signal from the DMA is typically used to start firmware execution, as noted in Table 4-33. Meanwhile, further DSP input data fills up the pong buffer.

When the pong input buffer is full, the DSP1_PONG_FULL bit is set, and another DSP start signal is generated. The DSP firmware must take care to read the input data from the applicable buffer, in accordance with the DSP1_PING_FULL and DSP1_PONG_FULL status bits.

Twin buffers are also used on the DSP output (RDMA) channels. The output ping buffers are emptied at the same time as the input ping buffers are filled; the output pong buffers are emptied at the same time that the input pong buffers are filled.

The DSP core supports 24-bit signal processing. Under default conditions, the DSP audio data is in 2's complement Q3.20 format (i.e., 0xF00000 corresponds to the -1.0 level, and 0x100000 corresponds to the +1.0 level; a sine wave with peak values of ± 1.0 corresponds to the 0 dBFS level). If DSP1_DMA_WORD_SEL is set, audio data is transferred to and from the DSP in Q0.23 format. The applicable format should be set according to the requirements of the specific DSP firmware.

Note that the DSP core is optimized for Q3.20 audio data processing; Q0.23 data can be supported, but the firmware implementation may incur a reduction in power efficiency due to the higher MIPS required for arithmetic operations in non-native data word format.

The DMA function is an input to the interrupt control circuit—see Section 4.4.5. The respective interrupt event is triggered if all enabled input (WDMA) channel buffers have been filled and all enabled output (RDMA) channel buffers have been emptied.

Further details of the DMA are provided in the software programming guide; contact your Cirrus Logic representative if required.



Register Address	Bit	Label	Default	Description
R1048068 (0xF_FE04)	31	DSP1_PING_FULL	0	DSP1 WDMA Ping Buffer Status
DSP1_Status_1				0 = Not Full
				1 = Full
	30	DSP1_PONG_FULL	0	DSP1 WDMA Pong Buffer Status
				0 = Not Full
				1 = Full
	23:16	DSP1_WDMA_ACTIVE_	0x00	DSP1 WDMA Channel Status
		CHANNELS[7:0]		There are eight WDMA channels; each bit of this field indicates
				the status of the respective WDMA channel.
				Each bit is coded as follows:
				0 = Inactive
	04.40		0.0000	
R1048080 (0xF_FE10)	31:16	DSP1_START_ADDRESS_ WDMA_BUFFER_1[15:0]	0x0000	DSP1 WDMA Channel 1 Start Address
DSP1_WDMA_Buffer_1				Bit [15] = Memory select
				0 = X-data memory
				1 = Y-data memory
				Bits [14:0] = Address select
				The address is defined relative to the base address of the applicable data memory. The LSB represents one 24-bit DSP
				memory word.
				Note that the start address is also controlled by the respective
				DSP1_WDMA_CHANNEL_OFFSET bit.
	15:0	DSP1_START_ADDRESS_	0x0000	DSP1 WDMA Channel 0 Start Address
		WDMA_BUFFER_0[15:0]		Field description is as above.
R1048082 (0xF_FE12)	31:16	DSP1_START_ADDRESS_	0x0000	DSP1 WDMA Channel 3 Start Address
DSP1_WDMA_Buffer_2		WDMA_BUFFER_3[15:0]		Field description is as above.
	15:0	DSP1 START ADDRESS	0x0000	DSP1 WDMA Channel 2 Start Address
		WDMA_BUFFER_2[15:0]		Field description is as above.
R1048084 (0xF_FE14)	31:16	DSP1_START_ADDRESS_	0x0000	DSP1 WDMA Channel 5 Start Address
DSP1_WDMA_Buffer_3		WDMA_BUFFER_5[15:0]		Field description is as above.
	15:0	DSP1_START_ADDRESS_	0x0000	DSP1 WDMA Channel 4 Start Address
		WDMA_BUFFER_4[15:0]		Field description is as above.
R1048086 (0xF_FE16)	31:16	DSP1_START_ADDRESS_	0x0000	DSP1 WDMA Channel 7 Start Address
DSP1_WDMA_Buffer_4		WDMA_BUFFER_7[15:0]		Field description is as above.
	15:0	DSP1_START_ADDRESS_	0x0000	DSP1 WDMA Channel 6 Start Address
		WDMA_BUFFER_6[15:0]		Field description is as above.
R1048096 (0xF_FE20)	31:16	DSP1_START_ADDRESS_	0x0000	DSP1 RDMA Channel 1 Start Address
DSP1_RDMA_Buffer_1		RDMA_BUFFER_1[15:0]		Bit [15] = Memory select
				0 = X-data memory
				1 = Y-data memory
				Bits [14:0] = Address select
				The address is defined relative to the base address of the
				applicable data memory. The LSB represents one 24-bit DSP
				memory word.
				Note that the start address is also controlled by the respective DSP1_RDMA_CHANNEL_OFFSET bit.
	15.0	DSP1 START ADDRESS	0x0000	DSP1_RDMA_CHANNEL_OFFSET bit.
	13.0	RDMA_BUFFER_0[15:0]	0,0000	Field description is as above.
R1048098 (0xF_FE22)	31.16	DSP1 START ADDRESS	0x0000	DSP1 RDMA Channel 3 Start Address
DSP1 RDMA Buffer 2	01.10	RDMA BUFFER 3[15:0]	0,0000	Field description is as above.
	15.0	DSP1 START ADDRESS	0x0000	DSP1 RDMA Channel 2 Start Address
	10.0	RDMA_BUFFER_2[15:0]	0,0000	Field description is as above.
R1048100 (0xF_FE24)	31.16	DSP1 START ADDRESS	0x0000	DSP1 RDMA Channel 5 Start Address
DSP1_RDMA_Buffer_3	01.10	RDMA_BUFFER_5[15:0]	0,0000	Field description is as above.
	15:0	DSP1 START ADDRESS	0x0000	DSP1 RDMA Channel 4 Start Address
	10.0	RDMA_BUFFER_4[15:0]	0,0000	Field description is as above.
		······································		

Table 4-34. DMA Control



Register Address	Bit	Label	Default	Description
R1048112 (0xF_FE30)	23:16	DSP1_WDMA_CHANNEL_	0x00	DSP1 WDMA Channel Enable
DSP1_DMA_Config_1		ENABLE[7:0]		There are eight WDMA channels; each bit of this field enables the respective WDMA channel.
				Each bit is coded as follows:
				0 = Disabled
				1 = Enabled
	13:0	DSP1_DMA_BUFFER_	0x0000	DSP1 DMA Buffer Length
		LENGTH[13:0]		Selects the amount of data transferred in each DMA channel. The LSB represents one 24-bit DSP memory word.
R1048114 (0xF_FE32)	7:0	DSP1_WDMA_CHANNEL_	0x00	DSP1 WDMA Channel Offset
DSP1_DMA_Config_2		OFFSET[7:0]		There are eight WDMA channels; each bit of this field offsets the start Address of the respective WDMA channel.
				Each bit is coded as follows:
				0 = No offset
				1 = Offset by 0x8000
R1048116 (0xF_FE34)	21:16	DSP1_RDMA_CHANNEL_	0x00	DSP1 RDMA Channel Offset
DSP1_DMA_Config_3		OFFSET[5:0]		There are six RDMA channels; each bit of this field offsets the start Address of the respective RDMA channel.
				Each bit is coded as follows:
				0 = No offset
				1 = Offset by 0x8000
	5:0	DSP1_RDMA_CHANNEL_	0x00	DSP1 RDMA Channel Enable
		ENABLE[5:0]		There are six RDMA channels; each bit of this field enables the respective RDMA channel.
				Each bit is coded as follows:
				0 = Disabled
				1 = Enabled
R1048118 (0xF_FE36)	0	DSP1_DMA_WORD_SEL	0	DSP1 Data Word Format
DSP1_DMA_Config_4				0 = Q3.20 format (4 integer bits, 20 fractional bits)
				1 = Q0.23 format (1 integer bit, 23 fractional bits)
				The data word format should be set according to the requirements of the applicable DSP firmware.

Table 4-34. DMA Control (Cont.)

4.4.5 DSP Interrupts

The DSP core provides inputs to the interrupt circuit and can be used to trigger an interrupt event when the associated conditions occur. The following interrupts are provided for DSP core:

- DMA interrupt—Asserted when all enabled DSP input (WDMA) channel buffers have been filled, and all enabled DSP output (RDMA) channel buffers have been emptied
- DSP Start 1, DSP Start 2 interrupts—Asserted when the respective start signal is triggered
- DSP Busy interrupt—Asserted when the DSP is busy (i.e., when firmware execution or DMA processes are started)
- DSP Bus Error interrupt—Asserted when a locked register address, invalid memory address, or watchdog timeout error is detected

The CS42L92 also provides 16 control bits that allow the DSP core to generate programmable interrupt events. When a 1 is written to these bits (see Table 4-35), the respective DSP interrupt (DSP_IRQn_EINTx) is triggered. The associated interrupt bits are latched once set; they can be polled at any time or used to control the IRQ signal.

See Section 4.15 for further details.

Register Address	Bit	Label	Default	Description
R5632 (0x1600)	1	DSP_IRQ2	0	DSP IRQ2. Write 1 to trigger the DSP_IRQ2_EINT <i>n</i> interrupt.
ADSP2_IRQ0	0	DSP_IRQ1	0	DSP IRQ1. Write 1 to trigger the DSP_IRQ1_EINT <i>n</i> interrupt.
R5633 (0x1601)	1	DSP_IRQ4	0	DSP IRQ4. Write 1 to trigger the DSP_IRQ4_EINT <i>n</i> interrupt.
ADSP2_IRQ1	0	DSP_IRQ3	0	DSP IRQ3. Write 1 to trigger the DSP_IRQ3_EINT <i>n</i> interrupt.

Table 4-35. DSP Interrupts



Register Address	Bit	Label	Default	Description
R5634 (0x1602)	1	DSP_IRQ6	0	DSP IRQ6. Write 1 to trigger the DSP_IRQ6_EINT <i>n</i> interrupt.
ADSP2_IRQ2	0	DSP_IRQ5	0	DSP IRQ5. Write 1 to trigger the DSP_IRQ5_EINT <i>n</i> interrupt.
R5635 (0x1603)	1	DSP_IRQ8	0	DSP IRQ8. Write 1 to trigger the DSP_IRQ8_EINT <i>n</i> interrupt.
ADSP2_IRQ3	0	DSP_IRQ7	0	DSP IRQ7. Write 1 to trigger the DSP_IRQ7_EINT <i>n</i> interrupt.
R5636 (0x1604)	1	DSP_IRQ10	0	DSP IRQ10. Write 1 to trigger the DSP_IRQ10_EINT <i>n</i> interrupt.
ADSP2_IRQ4	0	DSP_IRQ9	0	DSP IRQ9. Write 1 to trigger the DSP_IRQ9_EINT <i>n</i> interrupt.
R5637 (0x1605)	1	DSP_IRQ12	0	DSP IRQ12. Write 1 to trigger the DSP_IRQ12_EINT <i>n</i> interrupt.
ADSP2_IRQ5	0	DSP_IRQ11	0	DSP IRQ11. Write 1 to trigger the DSP_IRQ11_EINT <i>n</i> interrupt.
R5638 (0x1606)	1	DSP_IRQ14	0	DSP IRQ14. Write 1 to trigger the DSP_IRQ14_EINT <i>n</i> interrupt.
ADSP2_IRQ6	0	DSP_IRQ13	0	DSP IRQ13. Write 1 to trigger the DSP_IRQ13_EINT <i>n</i> interrupt.
R5639 (0x1607)	1	DSP_IRQ16	0	DSP IRQ16. Write 1 to trigger the DSP_IRQ16_EINT <i>n</i> interrupt.
ADSP2_IRQ7	0	DSP_IRQ15	0	DSP IRQ15. Write 1 to trigger the DSP_IRQ15_EINT <i>n</i> interrupt.

Table 4-35. DSP Interrupts (Cont.)

4.4.6 DSP Debug Support

General-purpose registers are provided for the DSP. These have no assigned function and can be used to assist in algorithm development.

The JTAG interface provides test and debug access to the CS42L92, as described in Section 4.20. The JTAG interface clock can be enabled for the DSP core using DSP1_DBG_CLK_ENA. Note that, when the JTAG interface is used to access the DSP core, the DSP1_CORE_ENA bit must also be set.

The DSP1_LOCK_ERR_STS bit indicates that the DSP attempted to write to a locked register address. The DSP1_ ADDR_ERR_STS bit indicates that the DSP attempted to access an invalid memory address (i.e., an address whose contents are undefined). Once set, these bits remain set until a 1 is written to DSP1_ERR_CLEAR.

The DSP1_PMEM_ERR_ADDR and DSP1_XMEM_ERR_ADDR fields contain the program memory and X-data memory addresses associated with a locked register address error condition. If DSP1_LOCK_ERR_STS is set, these fields correspond to the first-detected locked register address error. Note that no subsequent error event can be reported in these fields until the DSP1_LOCK_ERR_STS is cleared.

Note: The DSP1_PMEM_ERR_ADDR value is the prefetched address of a code instruction that has not yet been executed; it does not point directly to the instruction that caused the error.

The DSP1_BUS_ERR_ADDR field indicates the register/memory address that resulted in a register-access error. The field relates either to a locked register address error or to an invalid memory address error, as follows:

- If DSP1_LOCK_ERR_STS is set, the DSP1_BUS_ERR_ADDR value corresponds to the first-detected locked register address error. Note that no subsequent error event can be reported in this field until DSP1_LOCK_ERR_ STS is cleared.
- If DSP1_ADDR_ERR_STS is set, and DSP1_LOCK_ERR_STS is clear, the DSP1_BUS_ERR_ADDR field corresponds to the most recent invalid memory address error.
- If the DSP1_LOCK_ERR_STS and DSP1_ADDR_ERR_STS are both clear, the DSP1_BUS_ERR_ADDR field is undefined.
- **Note:** The DSP1_BUS_ERR_ADDR value is coded using a byte-referenced address, so the actual register address is equal to DSP1_BUS_ERR_ADDR / 2. If the register-access error is the result of an attempt to access the virtual DSP registers, a register address of 0 is reported.

If the DSP1_ERR_PAUSE bit is set, the DSP code execution stops immediately on detection of a locked register address error. This enables debug information to be retrieved from the DSP core during code development. In this event, code execution can be restarted by clearing the DSP1_ERR_PAUSE bit. Alternatively, the DSP core can restarted by clearing and setting DSP1_CORE_ENA (described in Section 4.4.3.3).



Register Address	Bit	Label	Default	Description
R1048064 (0xF_FE00)	3	DSP1_DBG_CLK_ENA	0	DSP1 Debug Clock Enable
DSP1_Config_1				0 = Disabled
				1 = Enabled
R1048128 (0xF_FE40)	31:16	DSP1_SCRATCH_1[15:0]	0x0000	DSP1 Scratch Register 1
DSP1_Scratch_1	15:0	DSP1_SCRATCH_0[15:0]	0x0000	DSP1 Scratch Register 0
R1048130 (0xF_FE42)	31:16	DSP1_SCRATCH_3[15:0]	0x0000	DSP1 Scratch Register 3
DSP1_Scratch_2	15:0	DSP1_SCRATCH_2[15:0]	0x0000	DSP1 Scratch Register 2
R1048146 (0xF_FE52)	23:0	DSP1_BUS_ERR_ADDR[23:0]	0x00_0000	Contains the register address of a memory region lock
DSP1_Bus_Error_Addr				or memory address error event.
				Note the associated register address is equal to DSP1_ BUS_ERR_ADDR / 2.
R1048186 (0xF_FE7A)	15	DSP1_LOCK_ERR_STS	0	DSP1 memory region lock error status.
DSP1_Region_lock_ctrl_0				This bit, when set, indicates that DSP1 attempted to write to a locked register address.
				This bit is latched when set; it is cleared when a 1 is written to DSP1_ERR_CLEAR.
	14	DSP1_ADDR_ERR_STS	0	DSP1 memory address error status.
				This bit, when set, indicates that DSP1 attempted to access an undefined locked register address.
				This bit is latched when set; it is cleared when a 1 is written to DSP1_ERR_CLEAR.
	1	DSP1_ERR_PAUSE	0	DSP1 bus address error control.
				Configures the DSP1 response to a memory region lock error event.
				0 = No action
				1 = Pause DSP1 code execution
	0	DSP1_ERR_CLEAR	0	Write 1 to clear the memory region lock error and memory address error status bits.
R1048188 (0xF_FE7C) DSP1_PMEM_Err_Addr XMEM_ERR_Addr	30:16	DSP1_PMEM_ERR_ADDR[14:0]	0x0000	Contains the program memory address of a memory region lock error event. Note this is the prefetched address of a subsequent instruction; it does not point directly to the address that caused the error.
	15:0	DSP1_XMEM_ERR_ADDR[15:0]	0x0000	Contains the X-data memory address of a memory region lock error event.

Table 4-36. DSP Debug Support

4.4.7 Virtual DSP Registers

The DSP control registers are described throughout Section 4.4. Each control register has a unique location within the CS42L92 register map.

An additional set of DSP control registers is also defined, which can be used in firmware to access the DSP control fields: the virtual DSP (or DSP 0) registers are defined at address R4096 (0x1000) in the device register map. The full register map listing is provided in Section 6.

Note that read/write access to the virtual DSP registers is only possible via firmware running on the integrated DSP core. When DSP firmware accesses the virtual registers, the registers are automatically mapped onto the DSP1 control registers. The virtual DSP registers are designed to allow software to be transferable across different DSPs (e.g., on multicore devices) without modification to the software code.

The virtual DSP registers are defined at register addresses R4096–R4192 (0x1000–0x1060) in the device register map. Note that these registers cannot be accessed directly at the addresses shown; they can be only accessed through DSP firmware code, using the register window function shown in Fig. 4-34. The virtual DSP registers are located at address 0xD000 in the X-data memory map.



4.5 DSP Peripheral Control

The CS42L92 incorporates a suite of DSP peripheral functions that can be integrated together to provide an enhanced capability for DSP applications. Configurable event log functions provide multichannel monitoring of internal and external signals. The general-purpose timer provides time-stamp data for the event logger; it also supports the watchdog and other miscellaneous time-based functions. Maskable GPIO provides an efficient mechanism for the DSP core to access the required input and output signals.

The DSP peripherals are designed to provide a comprehensive DSP capability, operating with a high degree of autonomy from the host processor.

4.5.1 Event Logger

The CS42L92 provides an event log function, supporting multichannel, edge-sensitive monitoring and recording of internal or external signals.

4.5.1.1 Overview

The event logger allows status information to be captured from a large number of sources, to be prioritized and acted upon as required. For the purposes of the event logger, an event is recorded when a logic transition (edge) is detected on a selected signal source.

The logged events are held in a FIFO buffer, which is managed by the application software. A 32-bit time stamp, derived from the general-purpose timer, is associated and recorded with each FIFO index, to provide a comprehensive record of the detected events.

The event logger must be associated with the general-purpose timer. The timer is the source of time stamp data for any logged events. If DSPCLK is disabled, the timer also provides the clock source for the event logger. (If DSPCLK is enabled, DSPCLK is used as the clock source instead.)

A maximum of one event per cycle of the clock source can be logged. If more than one event occurs within the cycle time, the highest priority (lowest channel number) event is logged at the rising edge of the clock. In this case, any lower priority events are queued, and are logged as soon as no higher priority events are pending. It is possible for recurring events on a high-priority channel to be logged, while low-priority ones remain queued. Note that recurring instances of queued events are not logged.

The event logger can use a slow clock (e.g., 32 kHz), but higher clock frequencies may also be commonly used, depending on the application and use case. The clock frequency determines the maximum possible event logging rate.

4.5.1.2 Event Logger Control

The event logger is enabled by setting EVENTLOG1_ENA. The event logger can be reset by writing 1 to EVENTLOG1_ RST—executing this function clears all the event logger status flags and clears the contents of the FIFO buffer.

The associated timer (and time-stamp source) is selected using EVENTLOG1_TIME_SEL. Note that the event logger must be disabled (EVENTLOG1_ENA = 0) when selecting the timer source.

4.5.1.3 Input Channel Configuration

The event logger allows up to 16 input channels to be configured for detection and logging. The EVENTLOG1_CHx_SEL field selects the applicable input source for each channel (where *x* identifies the channel number, 1 to 16). The polarity selection and debounce options are configured using the EVENTLOG1_CHx_POL and EVENTLOG1_CHx_DB bits respectively.

To avoid filling the FIFO buffer with repeated instances of any event, a selectable filter is provided for each input channel. If the EVENTLOG1_CHx_FILT bit is set, new events on the respective channel are ignored by the event logger if the unread entries in the FIFO buffer indicate a previous event of the same type (i.e., same input source and same polarity). The read/write pointers of the FIFO buffer (see Section 4.5.1.4) are used to determine which FIFO entries are unread (i.e., have not yet been read by the host processor).



The input channels can be enabled or disabled freely, using EVENTLOG1_CHx_ENA, without having to disable the event logger entirely. An input channel must be disabled whenever the associated x_SEL , x_FILT , x_POL , or x_DB fields are written. It is possible to reconfigure input channels while the event logger is enabled, provided the channels being reconfigured are disabled when doing so.

The available input sources include GPIO inputs, external accessory status (jack, mic, sensors), and signals generated by the integrated DSP core. A list of the valid input sources for the event logger is provided in Table 4-38. Note that, to log both rising and falling events from any source, two separate input channels must be configured—one for each polarity.

If an input channel is configured for rising edge detection (EVENTLOG1_CHx_POL = 0), and the corresponding input signal is asserted (Logic 1) at the time when the event logger is enabled, an event is logged in respect of this initial state. Similarly, if an input channel is configured for falling edge detection, and is deasserted (Logic 0) when the event logger is enabled, a corresponding event is logged. If rising and falling edges are both configured for detection, an event is always logged in respect of the initial condition.

4.5.1.4 FIFO Buffer

Each event (signal transition) that meets the criteria of an enabled channel is written to the 16-stage FIFO buffer. The buffer is filled cyclically, but does not overwrite unread data when full. A status bit is provided to indicate if the buffer fills up completely.

Note that the FIFO behavior is not enforced or fully implemented in the device hardware, but assumes that a compatible software implementation is in place. New events are written to the buffer in a cyclic manner, but the data can be read out in any order, if desired. The designed FIFO behavior requires the software to update the read pointer (RPTR) in the intended manner for smooth operation.

The entire contents of the 16-stage FIFO buffer can be accessed directly in the register map. Each FIFO index (y = 0 to 15) comprises the EVENTLOG1_FIFOy_ID (identifying the source signal of the associated log event), the EVENTLOG1_FIFOy_POL (the polarity of the respective event transition), and the EVENTLOG1_FIFOy_TIME field (containing the 32-bit time stamp from the timer).

The FIFO buffer is managed using EVENTLOG1_FIFO_WPTR and EVENTLOG1_FIFO_RPTR. The write pointer (WPTR) field identifies the index location (0 to 15) in which the next event is logged. The read pointer (RPTR) field identifies the index location of the first set of unread data, if any exists. Both of these fields are initialized to 0 when the event logger is reset.

If RPTR ≠ WPTR, the buffer contains new data. The number of new events is equal to the difference between the two pointer values (WPTR – RPTR, allowing for wraparound beyond Index 15). For example, if WPTR = 12 and RPTR = 8, this means that there are four unread data sets in the buffer, at index locations 8, 9, 10, and 11.

After reading the new data from the buffer, the RPTR value should be incremented by the corresponding amount (e.g., increment by 4, in the example described above). Note that the RPTR value can either be incremented once for each read, or can be incremented in larger steps after a batch read.

• If RPTR = WPTR, the buffer is either empty (0 events) or full (16 events). In this case, the status bits described in Section 4.5.1.5 confirm the current status of the buffer.

4.5.1.5 Status Bits

The EVENTLOG1_CHx_STS bits indicate the status of the source signal for the respective input channel. Note that the status indication is not valid for all input source selections—it is not possible to provide indication of transitory events (e.g., microphone accessory detection).

The EVENTLOG1_NOT_EMPTY bit indicates whether the FIFO buffer is empty. If this bit is set, it indicates one or more new sets of data in the FIFO.

The EVENTLOG1_WMARK_STS bit indicates when the number of FIFO index locations available for new events reaches a configurable threshold, known as the watermark level. The watermark level is held in the EVENTLOG1_FIFO_WMARK field.



The EVENTLOG1_FULL bit indicates when the FIFO buffer is full. If this bit is set, it indicates that there are 16 sets of new event data in the FIFO. Note that this does not mean that a buffer overflow condition has occurred, but further events are not logged or indicated until the buffer has been cleared.

Note: Following a buffer full condition, the FIFO operation resumes as soon as the RPTR field has been updated to a new value. Writing the same value to RPTR does not restart the FIFO operation, even if the entire buffer contents have been read. After all of the required data has been read from the buffer, the RPTR value should be set equal to the WPTR value; an intermediate (different) value must also be written to the RPTR field in order to clear the buffer full status and restart the FIFO operation.

4.5.1.6 Interrupts, GPIO, Write Sequencer, and DSP Firmware Control

The control-write sequencer is automatically triggered whenever the NOT_EMPTY status of the event log buffer is asserted. See Section 4.18 for further details.

The event log status flags are inputs to the interrupt control circuit and can be used to trigger an interrupt event when the respective FIFO condition (full, not empty, or watermark level) occurs; see Section 4.15.

The event log status can be output directly on a GPIO pin as an external indication of the event logger; see Section 4.14 to configure a GPIO pin for this function.

The event log NOT_EMPTY status can also be selected as a start trigger for DSP firmware execution; see Section 4.4.

4.5.1.7 Event Logger Control Registers

The event logger control registers are described in Table 4-37.

Register Address	Bit	Label	Default	Description
R294912 (0x4_8000)	1	EVENTLOG1_RST	0	Event Log Reset
EVENTLOG1 CONTROL				Write 1 to reset the status outputs and clear the FIFO buffer.
_	0	EVENTLOG1_ENA	0	Event Log Enable
				0 = Disabled
				1 = Enabled
R294916 (0x4_8004)	1:0	EVENTLOG1_TIMER_	00	Event Log Timer Source Select
EVENTLOG1_TIMER_SEL		SEL[1:0]		00 = Timer 1
				All other codes are reserved. Note that the event log must be disabled when updating this field.
R294924 (0x4_800C) EVENTLOG1_FIFO_ CONTROL1	3:0	EVENTLOG1_FIFO_ WMARK[3:0]	0x1	Event Log FIFO Watermark. The watermark status output is asserted when the number of FIFO locations available for new events is less than or equal to the FIFO watermark.
				Valid from 0 to 15.
R294926 (0x4_800E) EVENTLOG1_FIFO_ POINTER1	18	EVENTLOG1_FULL	0	Event Log FIFO Full Status. This bit, when set, indicates that the FIFO buffer is full. It is cleared when a new value is written to the FIFO read pointer, or when the event log is Reset.
	17	EVENTLOG1_WMARK_STS	0	Event Log FIFO Watermark Status. This bit, when set, indicates that the FIFO space available for new events to be logged is less than or equal to the watermark threshold.
	16	EVENTLOG1_NOT_EMPTY	0	Event Log FIFO Not-Empty Status. This bit, when set, indicates one or more new sets of logged event data in the FIFO.
	11:8	EVENTLOG1_FIFO_ WPTR[3:0]	0x0	Event Log FIFO Write Pointer. Indicates the FIFO index location in which the next event is logged. This is a read-only field.
	3:0	EVENTLOG1_FIFO_ RPTR[3:0]	0x0	Event Log FIFO Read Pointer. Indicates the FIFO index location of the first set of unread data, if any exists. For the intended FIFO behavior, this field must be incremented after the respective data has been read.

Table 4-37. Event Logger (EVENTLOG1) Control



Register Address	Bit	Label	Default	Description
R294944 (0x4_8020)	15	EVENTLOG1_CH16_ENA	0	Event Log Channel 16 Enable
EVENTLOG1_CH_ENABLE				0 = Disabled, 1 = Enabled
	14	EVENTLOG1_CH15_ENA	0	Event Log Channel 15 Enable
				0 = Disabled, 1 = Enabled
	13	EVENTLOG1_CH14_ENA	0	Event Log Channel 14 Enable
				0 = Disabled, 1 = Enabled
	12	EVENTLOG1_CH13_ENA	0	Event Log Channel 13 Enable
				0 = Disabled, 1 = Enabled
	11	EVENTLOG1_CH12_ENA	0	Event Log Channel 12 Enable
				0 = Disabled, 1 = Enabled
	10	EVENTLOG1_CH11_ENA	0	Event Log Channel 11 Enable
				0 = Disabled, 1 = Enabled
	9	EVENTLOG1_CH10_ENA	0	Event Log Channel 10 Enable
				0 = Disabled, 1 = Enabled
	8	EVENTLOG1_CH9_ENA	0	Event Log Channel 9 Enable
				0 = Disabled, 1 = Enabled
	7	EVENTLOG1_CH8_ENA	0	Event Log Channel 8 Enable
				0 = Disabled, 1 = Enabled
	6	EVENTLOG1_CH7_ENA	0	Event Log Channel 7 Enable
				0 = Disabled, 1 = Enabled
	5	EVENTLOG1_CH6_ENA	0	Event Log Channel 6 Enable
				0 = Disabled, 1 = Enabled
	4	EVENTLOG1_CH5_ENA	0	Event Log Channel 5 Enable
				0 = Disabled, 1 = Enabled
	3	EVENTLOG1_CH4_ENA	0	Event Log Channel 4 Enable
				0 = Disabled, 1 = Enabled
	2	EVENTLOG1_CH3_ENA	0	Event Log Channel 3 Enable
				0 = Disabled, 1 = Enabled
	1	EVENTLOG1_CH2_ENA	0	Event Log Channel 2 Enable
				0 = Disabled, 1 = Enabled
	0	EVENTLOG1_CH1_ENA	0	Event Log Channel 1 Enable
				0 = Disabled, 1 = Enabled
R294948 (0x4_8024)		EVENTLOG1_CH16_STS	0	Event Log Channel 16 Status
EVENTLOG1_CH_STATUS		EVENTLOG1_CH15_STS	0	Event Log Channel 15 Status
		EVENTLOG1_CH14_STS	0	Event Log Channel 14 Status
		EVENTLOG1_CH13_STS	0	Event Log Channel 13 Status
		EVENTLOG1_CH12_STS	0	Event Log Channel 12 Status
		EVENTLOG1_CH11_STS	0	Event Log Channel 11 Status
		EVENTLOG1_CH10_STS	0	Event Log Channel 10 Status
	8	EVENTLOG1_CH9_STS	0	Event Log Channel 9 Status
	7	EVENTLOG1_CH8_STS	0	Event Log Channel 8 Status
	6	EVENTLOG1_CH7_STS	0	Event Log Channel 7 Status
		EVENTLOG1_CH6_STS	0	Event Log Channel 6 Status
		EVENTLOG1_CH5_STS	0	Event Log Channel 5 Status
		EVENTLOG1_CH4_STS	0	Event Log Channel 4 Status
	2	EVENTLOG1_CH3_STS	0	Event Log Channel 3 Status
	1	EVENTLOG1_CH2_STS	0	Event Log Channel 2 Status
	0	EVENTLOG1_CH1_STS	0	Event Log Channel 1 Status

Table 4-37. Event Logger (EVENTLOG1) Control (Cont.)



Register Address	Bit	Label	Default	Description
R294976 (0x4_8040)	15	EVENTLOG1_CHn_DB	0	Event Log Channel <i>n</i> debounce
EVENTLOG1_CH1_DEFINE				0 = Disabled, 1 = Enabled
to				Note that channel must be disabled when updating this field.
R295006 (0x4_805E)	14	EVENTLOG1_CHn_POL	0	Event Log Channel <i>n</i> polarity
EVENTLOG1 CH16 DEFINE				0 = Rising edge triggered, 1 = Falling edge triggered
				Note that channel must be disabled when updating this field.
	13	EVENTLOG1 CHn FILT	0	Event Log Channel <i>n</i> filter
	_		-	0 = Disabled, 1 = Enabled
				If the filter is enabled, the channel is ignored if the FIFO
				contains unread events of the same source/polarity.
				Note that channel must be disabled when updating this field.
	9:0	EVENTLOG1_CHn_SEL[9:0]	0x000	Event Log Channel <i>n</i> source ¹
				Note that channel must be disabled when updating this field.
R295040 (0x4 8080)	12	EVENTLOG1 FIFO0 POL	0	Event Log FIFO Index 0 polarity
EVENTLOG1 FIFO0 READ				0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1 FIFO0 ID[9:0]	0x000	Event Log FIFO Index 0 source 1
R295042 (0x4 8082)		EVENTLOG1 FIFO0		Event Log FIFO Index 0 Time
EVENTLOG1 FIFO0 TIME		TIME[31:0]	_0000	
R295044 (0x4 8084)	12	EVENTLOG1 FIFO1 POL	0	Event Log FIFO Index 1 polarity
EVENTLOG1 FIFO1 READ			Ŭ	0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1 FIFO1 ID[9:0]	0x000	Event Log FIFO Index 1 source 1
R295046 (0x4 8086)	31:0	EVENTLOG1 FIF01		Event Log FIFO Index 1 Time
EVENTLOG1 FIFO1 TIME	51.0	TIME[31:0]	_0000	Event Log I II O Index I Time
R295048 (0x4 8088)	12	EVENTLOG1 FIFO2 POL	0	Event Log FIFO Index 2 polarity
EVENTLOG1 FIFO2 READ	12		0	0 = Rising edge, 1 = Falling edge
EVENTLOGT_FIFO2_READ	0.0	EVENTLOG1 FIFO2 ID[9:0]	0x000	Event Log FIFO Index 2 source ¹
				•
R295050 (0x4_808A)	31:0	EVENTLOG1_FIFO2_ TIME[31:0]	0x0000 _0000	Event Log FIFO Index 2 Time
EVENTLOG1_FIFO2_TIME	40			Event Lan EIEO Index 2 a clarity
R295052 (0x4_808C)	12	EVENTLOG1_FIFO3_POL	0	Event Log FIFO Index 3 polarity
EVENTLOG1_FIFO3_READ				0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1_FIFO3_ID[9:0]		Event Log FIFO Index 3 source 1
R295054 (0x4_808E)	31:0	EVENTLOG1_FIFO3_		Event Log FIFO Index 3 Time
EVENTLOG1_FIFO3_TIME		TIME[31:0]	_0000	
R295056 (0x4_8090)	12	EVENTLOG1_FIFO4_POL	0	Event Log FIFO Index 4 polarity
EVENTLOG1_FIFO4_READ				0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1_FIFO4_ID[9:0]		Event Log FIFO Index 4 source ¹
R295058 (0x4_8092)	31:0	EVENTLOG1_FIFO4_	0x0000	Event Log FIFO Index 4 Time
EVENTLOG1_FIFO4_TIME		TIME[31:0]	_0000	
R295060 (0x4_8094)	12	EVENTLOG1_FIFO5_POL	0	Event Log FIFO Index 5 polarity
EVENTLOG1_FIFO5_READ				Field description is as above.
	9:0	EVENTLOG1_FIFO5_ID[9:0]	0x000	Event Log FIFO Index 5 source ¹
R295062 (0x4_8096)	31:0	EVENTLOG1_FIFO5_		Event Log FIFO Index 5 Time
EVENTLOG1_FIFO5_TIME		TIME[31:0]	_0000	
R295064 (0x4_8098)	12	EVENTLOG1_FIFO6_POL	0	Event Log FIFO Index 6 polarity
EVENTLOG1_FIFO6_READ				0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1_FIFO6_ID[9:0]	0x000	Event Log FIFO Index 6 source 1
R295066 (0x4_809A)	31:0	EVENTLOG1_FIFO6_	0x0000	Event Log FIFO Index 6 Time
EVENTLOG1_FIFO6_TIME		TIME[31:0]	_0000	
R295068 (0x4_809C)	12	EVENTLOG1_FIFO7_POL	0	Event Log FIFO Index 7 polarity
EVENTLOG1_FIF07_READ				0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1_FIFO7_ID[9:0]	0x000	Event Log FIFO Index 7 source ¹
R295070 (0x4_809E)	31:0	EVENTLOG1_FIF07_		Event Log FIFO Index 7 Time
EVENTLOG1_FIFO7_TIME		TIME[31:0]	_0000	J
R295072 (0x4_80A0)	12	EVENTLOG1_FIF08_POL	0	Event Log FIFO Index 8 polarity
EVENTLOG1_FIF08_READ			Ĭ	0 = Rising edge, 1 = Falling edge
	9:0	EVENTLOG1 FIFO8 ID[9:0]	0x000	Event Log FIFO Index 8 source ¹
	0.0		0,000	

Table 4-37. Event Logger (EVENTLOG1) Control (Cont.)



Register Address	Bit	Label	Default	Description
R295074 (0x4_80A2)		EVENTLOG1_FIF08_		Event Log FIFO Index 8 Time
EVENTLOG1_FIF08_TIME	01.0	TIME[31:0]	_0000	
R295076 (0x4_80A4)	12	EVENTLOG1_FIFO9_POL	0	Event Log FIFO Index 9 polarity
EVENTLOG1_FIFO9_READ				0 = Rising edge, 1 = Falling edge
		EVENTLOG1_FIFO9_ID[9:0]	0x000	Event Log FIFO Index 9 source ¹
R295078 (0x4_80A6)	31:0	EVENTLOG1_FIFO9_		Event Log FIFO Index 9 Time
EVENTLOG1_FIFO9_TIME		TIME[31:0]	_0000	
R295080 (0x4_80A8)	12	EVENTLOG1_FIFO10_POL	0	Event Log FIFO Index 10 polarity
EVENTLOG1_FIFO10_READ				0 = Rising edge, 1 = Falling edge
		EVENTLOG1_FIFO10_ID[9:0]		Event Log FIFO Index 10 source ¹
R295082 (0x4_80AA)	31:0	EVENTLOG1_FIFO10_	0x0000	Event Log FIFO Index 10 Time
EVENTLOG1_FIFO10_TIME		TIME[31:0]	_0000	
R295084 (0x4_80AC)	12	EVENTLOG1_FIFO11_POL	0	Event Log FIFO Index 11 polarity
EVENTLOG1_FIFO11_READ				0 = Rising edge, 1 = Falling edge
		EVENTLOG1_FIFO11_ID[9:0]		Event Log FIFO Index 11 source ¹
R295086 (0x4_80AE)	31:0	EVENTLOG1_FIFO11_		Event Log FIFO Index 11 Time
EVENTLOG1_FIFO11_TIME		TIME[31:0]	_0000	
R295088 (0x4_80B0)	12	EVENTLOG1_FIFO12_POL	0	Event Log FIFO Index 12 polarity
EVENTLOG1_FIFO12_READ				0 = Rising edge, 1 = Falling edge
		EVENTLOG1_FIFO12_ID[9:0]		Event Log FIFO Index 12 source 1
R295090 (0x4_80B2)	31:0	EVENTLOG1_FIFO12_	0x0000	Event Log FIFO Index 12 Time
EVENTLOG1_FIFO12_TIME		TIME[31:0]	_0000	
R295092 (0x4_80B4)	12	EVENTLOG1_FIFO13_POL	0	Event Log FIFO Index 13 polarity
EVENTLOG1_FIFO13_READ				0 = Rising edge, 1 = Falling edge
		EVENTLOG1_FIFO13_ID[9:0]		Event Log FIFO Index 13 source ¹
R295094 (0x4_80B6)	31:0	EVENTLOG1_FIFO13_ TIME[31:0]	0x0000 _0000	Event Log FIFO Index 13 Time
EVENTLOG1_FIFO13_TIME	40			
R295096 (0x4_80B8)	12	EVENTLOG1_FIFO14_POL	0	Event Log FIFO Index 14 polarity
EVENTLOG1_FIFO14_READ	0.0		0,000	0 = Rising edge, 1 = Falling edge
		EVENTLOG1_FIFO14_ID[9:0] EVENTLOG1_FIFO14		Event Log FIFO Index 14 source ¹
R295098 (0x4_80BA)	31:0	TIME[31:0]	0x0000 _0000	Event Log FIFO Index 14 Time
EVENTLOG1_FIFO14_TIME	12	EVENTLOG1 FIFO15 POL	_0000	Event Log FIFO Index 15 polarity
R295100 (0x4_80BC)	12		U	0 = Rising edge, 1 = Falling edge
EVENTLOG1_FIFO15_READ	9:0	EVENTLOG1_FIFO15_ID[9:0]	0x000	Event Log FIFO Index 15 source ¹
R295102 (0x4_80BE)		EVENTLOG1_FIF015_ID[9.0]		Event Log FIFO Index 15 Source
EVENTLOG1 FIFO15 TIME	51.0	TIME[31:0]	_0000	LVEIL LOGT IL O IIIGEN 15 TIME
		-[]		

Table 4-37. Event Logger (EVENTLOG1) Control (Cont.)

1. See Table 4-38 for valid channel source selections

4.5.1.8 Event Logger Input Sources

A list of the valid input sources for the event logger is provided in Table 4-38.

The EDGE type noted is coded as *S* (single edge) or *D* (dual edge). Note that a single-edge input source only provides valid input to the event logger in the default (rising edge triggered) polarity.

Take care when enabling IRQ1 or IRQ2 as an input source for the event logger; a recursive loop, where the IRQ*n* signal is also an output from the event logger, must be avoided.

ID	Description	Edge
3	irq1	D
4	irq2	D
9	sysclk_fail	S
24	fll1_lock	D
25	fll2 lock	D

Table 4-38. Event Logger Input Sources

ID	Description	Edge
137	asrc1_in2_lock	D
160	dsp_irq1	S
161	dsp_irq2	S
162	dsp_irq3	S
163	dsp_irq4	S

	ID	Description	Edge
ſ	258	gpio3	D
ľ	259	gpio4	D
	260	gpio5	D
Ĭ	261	gpio6	D
l	262	gpio7	D



ID	Description	Edge	ID	Description	Edge	ID	Description	Edge
28	sysclk_err	D	164	dsp_irq5	S	263	gpio8	D
29	asyncclk_err	D	165	dsp_irq6	S	264	gpio9	D
30	dspclk_err	D	166	dsp_irq7	S	265	gpio10	D
32	frame_start_g1r1	S	167	dsp_irq8	S	266	gpio11	D
33	frame_start_g1r2	S	168	dsp_irq9	S	267	gpio12	D
34	frame_start_g1r3	S	169	dsp_irq10	S	268	gpio13	D
40	frame_start_g2r1_sys	S	170	dsp_irq11	S	269	gpio14	D
41	frame_start_g2r2_sys	S	171	dsp_irq12	S	270	gpio15	D
80	hpdet	S	172	dsp_irq13	S	271	gpio16	D
88	micdet1	S	173	dsp_irq14	S	320	Timer1	S
89	micdet2	S	174	dsp_irq15	S	336	event1_not_empty	S
96	jd1_rise	S	175	dsp_irq16	S	352	event1_full	S
97	jd1_fall	S	176	hp1l_sc	S	368	event1_wmark	S
98	jd2_rise	S	177	hp1r_sc	S	384	dsp1_dma	S
99	jd2_fall	S	178	hp2l_sc	S	416	dsp1_start1	S
100	micd_clamp_rise	S	179	hp2r_sc	S	432	dsp1_start2	S
101	micd_clamp_fall	S	180	hp3l_sc	S	448	dsp1_start	S
104	jd3_rise	S	181	hp3r_sc	S	464	dsp1_busy	D
105	jd3_fall	S	184	hp4l_sc	S	512	dsp1_bus_err	S
128	drc1_sig_det	D	185	hp4r_sc	S	560	alarm1_ch1	S
129	drc2_sig_det	D	236	dfc_saturate	S	561	alarm1_ch2	S
130	inputs_sig_det	D	256	gpio1	D	562	alarm1_ch3	S
136	asrc1_in1_lock	D	257	gpio2	D	563	alarm1_ch4	S

Table 4-38. Event Logger Input Sources (Cont.)

4.5.2 Alarm Generators

The CS42L92 provides four alarm-generator circuits are associated with the general-purpose timer. These can be used to generate interrupt events according to the count value of the timer. The alarm interrupts can be either one-off events, or can be configured for cyclic (repeated) triggers.

4.5.2.1 Alarm Control

The alarm is enabled by writing 1 to the ALM1_CH*n*_START bit (where *n* identifies the respective alarm, 1–4). The alarm is disabled by writing 1 to ALM1_CH*n*_STOP.

The operating mode of each alarm is configured using ALM1_CH*n*_TRIG_MODE. In each mode, the alarm events are controlled by the alarm-trigger value, ALM1_CH*n*_TRIG_VAL.

- In Absolute Mode, the alarm output is triggered when the timer count value is equal to the alarm trigger value.
- In Relative Mode, the alarm output is triggered when the timer count value has incremented by a number equal to the alarm trigger value—this mode counts the number of clock cycles after the ALM1_CHn_START bit is written.
- In Combination Mode, the alarm output is initially triggered as described for the Absolute Mode; the alarm then operates as described for the Relative Mode.

When the alarm output is triggered, an output signal is asserted for the respective alarm. The output is asserted for a duration that is configured using ALM1_CH*n*_PULSE_DUR. The resulting signal can be output directly on a GPIO pin.

If an alarm is enabled and an update is written to ALM1_CH*n*_TRIG_VAL or ALM1_CH*n*_PULSE_DUR, the new value is loaded into the respective control register, but does not reconfigure the alarm immediately. If the ALM1_CH*n*_UPD bit is set, the alarm-trigger and pulse-duration values are updated when the alarm is next triggered. The alarm-trigger and pulse-duration settings can also be updated by writing 1 to ALM1_CH*n*_START.

Note that, if an alarm is enabled, the general-purpose timer must be configured for continuous, count-up operation. The TIMER1_MAX_COUNT value must be greater than the respective ALM1_CH*n*_TRIG_VAL setting.



4.5.2.2 Interrupts and GPIO Output

The alarm generators provide input to the interrupt control circuit and can be used to trigger an interrupt event when the alarm-trigger conditions are met. An interrupt event is triggered on the rising edge of the alarm output signal.

The alarm output status bits, TIMER_ALM1_CH*n*_STS*x*, are asserted for a duration that is configured using ALM1_CH*n*_PULSE_DUR. Note that the TIMER_ALM1_CH*n*_STS1 and TIMER_ALM1_CH*n*_STS2 bits provide the same information.

See Section 4.15 for details of the CS42L92 interrupt controller.

The alarm status can be output directly on a GPIO pin as an external indication of the alarm events. See Section 4.14 to configure a GPIO pin for this function.

4.5.2.3 Alarm Control Registers

The alarm control registers are described in Table 4-39.

Register Address	Bit	Label	Default	Description
R303104 (0x4A000)	0	ALM1_TIMER_	0	Alarm block ALM1 timer source select
ALM1_CFG		SEL		0 = Timer 1
				All other codes are reserved.
				All ALM1 channels must be disabled when updating this register.
R303120 (0x4A010)	4	ALM1_CH1_	0	Channel 1 continuous mode select
ALM1_CONFIG1		CONT		0 = Single mode
				1 = Continuous mode
				Channel 1 must be disabled (ALM1_CH1_STS = 0) when updating this field.
	1:0	ALM1_CH1_	00	Channel 1 trigger mode select
		TRIG_MODE[1:0]		00 = Absolute Mode: Alarm is triggered when the count value of the timer source is equal to ALM1_CH1_TRIG_VAL.
				01 = Relative Mode: Alarm is triggered when the count value has incremented by a number equal to ALM1_CH1_TRIG_VAL.
				10 = Combination Mode: Alarm is initially triggered as described for Absolute Mode; the alarm then operates as described for Relative Mode.
				11 = Reserved
				Channel 1 must be disabled (ALM1_CH1_STS = 0) when updating this field.
R303122 (0x4A012) ALM1 CTRL1	15	ALM1_CH1_UPD	0	Channel 1 update control—Write 1 to indicate a new trigger value or pulse duration is ready to be applied.
				If Channel 1 is enabled and ALM1_CH1_UPD is set, the ALM1_CH1_TRIG_VAL and ALM1_CH1_PULSE_DUR settings are updated when the alarm is next triggered or by writing 1 to ALM1 CH1 START.
				If Channel 1 is disabled, the ALM1_CH1_UPD bit has no effect, and the ALM1_ CH1_TRIG_VAL and ALM1_CH1_PULSE_DUR settings are updated immediately when writing to the respective fields.
	4	ALM1 CH1		Channel 1 stop control—Write 1 to disable Channel 1
	-	STOP		
	0	ALM1_CH1_ START	—	Channel 1 start control—Write 1 to enable or restart Channel 1
R303124 (0x4A014)	31:0	ALM1_CH1_		Channel 1 alarm trigger value
ALM1_TRIG_VAL1		TRIG_VAL[31:0]	_0000	
R303126 (0x4A016)	31:0	ALM1_CH1_	0x0000	Channel 1 alarm output pulse duration
ALM1_PULSE_DUR1		PULSE_ DUR[31:0]	_0000	The pulse duration is referenced to the count rate of the selected timer source
R303128 (0x4A018)	0	ALM1_CH1_STS	0	Channel 1 status
ALM1_STATUS1				0 = Disabled
				1 = Enabled

Table 4-39. Alarm (ALM1) Control



Register Address	Bit	Label	Default	Description
R303136 (0x4A020)	4	ALM1_CH2_	0	Channel 2 continuous mode select
ALM1_CONFIG2		CONT		0 = Single mode
				1 = Continuous mode
				Channel 2 must be disabled (ALM1_CH2_STS = 0) when updating this field.
	1:0	ALM1_CH2_	00	Channel 2 trigger mode select
		TRIG_MODE[1:0]		00 = Absolute Mode: Alarm is triggered when the count value of the timer source is equal to ALM1_CH2_TRIG_VAL.
				01 = Relative Mode: Alarm is triggered when the count value has incremented by a number equal to ALM1_CH2_TRIG_VAL.
				10 = Combination Mode: Alarm is initially triggered as described for Absolute Mode; the alarm then operates as described for Relative Mode.
				11 = Reserved
				Channel 2 must be disabled (ALM1_CH2_STS = 0) when updating this field.
R303138 (0x4A022)	15	ALM1 CH2 UPD	0	Channel 2 update control—Write 1 to indicate a new trigger value or pulse duration
ALM1_CTRL2				is ready to be applied.
				If Channel 2 is enabled and ALM1_CH2_UPD is set, the ALM1_CH2_TRIG_VAL and ALM1_CH2_PULSE_DUR settings are updated when the alarm is next triggered or by writing 1 to ALM1_CH2_START.
				If Channel 2 is disabled, the ALM1_CH2_UPD bit has no effect, and the ALM1_ CH2_TRIG_VAL and ALM1_CH2_PULSE_DUR settings are updated immediately when writing to the respective fields.
	4	ALM1 CH2		Channel 2 stop control—Write 1 to disable Channel 2
	•	STOP		
	0	ALM1_CH2_ START	—	Channel 2 start control—Write 1 to enable or restart Channel 2
R303140 (0x4A024) ALM1_TRIG_VAL2	31:0	ALM1_CH2_ TRIG_VAL[31:0]	0x0000 _0000	Channel 2 alarm trigger value
R303142 (0x4A026)	31:0	ALM1 CH2	0x0000	Channel 2 alarm output pulse duration
ALM1_PULSE_DUR2		PULSE DUR[31:0]	_0000	The pulse duration is referenced to the count rate of the selected timer source
R303144 (0x4A028)	0	ALM1_CH2_STS	0	Channel 2 status
ALM1_STATUS2				0 = Disabled
				1 = Enabled
R303152 (0x4A030)	4	ALM1_CH3_	0	Channel 3 continuous mode select
ALM1_CONFIG3		CONT		0 = Single mode
				1 = Continuous mode
				Channel 3 must be disabled (ALM1_CH3_STS = 0) when updating this field.
	1:0	ALM1_CH3_	00	Channel 3 trigger mode select
		TRIG_MODE[1:0]		00 = Absolute Mode: Alarm is triggered when the count value of the timer source is equal to ALM1 CH3 TRIG VAL.
				01 = Relative Mode: Alarm is triggered when the count value has incremented by a number equal to ALM1_CH3_TRIG_VAL.
				10 = Combination Mode: Alarm is initially triggered as described for Absolute Mode; the alarm then operates as described for Relative Mode.
				11 = Reserved
				Channel 3 must be disabled (ALM1_CH3_STS = 0) when updating this field.
R303154 (0x4A032)	15	ALM1_CH3_UPD	0	Channel 3 update control—Write 1 to indicate a new trigger value or pulse duration is ready to be applied.
ALM1_CTRL3				If Channel 3 is enabled and ALM1_CH3_UPD is set, the ALM1_CH3_TRIG_VAL
				and ALM1_CH3_PULSE_DUR settings are updated when the alarm is next triggered or by writing 1 to ALM1_CH3_START.
				If Channel 3 is disabled, the ALM1_CH3_UPD bit has no effect, and the ALM1_ CH3_TRIG_VAL and ALM1_CH3_PULSE_DUR settings are updated
				immediately when writing to the respective fields.
	4	ALM1_CH3_ STOP	_	Channel 3 stop control—Write 1 to disable Channel 3
	0	ALM1_CH3_ START	—	Channel 3 start control—Write 1 to enable or restart Channel 3
R303156 (0x4A034)	31:0	ALM1_CH3_	0x0000	Channel 3 alarm trigger value
ALM1_TRÌG_VAL3		TRIG_VAL[31:0]	_0000	

Table 4-39. Alarm (ALM1) Control (Cont.)



Register Address	Bit	Label	Default	Description
R303158 (0x4A036)	31:0	ALM1_CH3_		Channel 3 alarm output pulse duration
ALM1_PULSE_DUR3		PULSE DUR[31:0]	_0000	The pulse duration is referenced to the count rate of the selected timer source
R303160 (0x4A038)	0	ALM1_CH3_STS	0	Channel 3 status
ALM1_STATUS3				0 = Disabled
				1 = Enabled
R303168 (0x4A040)	4	ALM1_CH4_	0	Channel 4 continuous mode select
ALM1_CONFIG4		CONT		0 = Single mode
				1 = Continuous mode
				Channel 4 must be disabled (ALM1_CH4_STS = 0) when updating this field.
	1:0	ALM1_CH4_	00	Channel 4 trigger mode select
		TRIG_MODE[1:0]		00 = Absolute Mode: Alarm is triggered when the count value of the timer source is equal to ALM1_CH4_TRIG_VAL.
				01 = Relative Mode: Alarm is triggered when the count value has incremented by a number equal to ALM1_CH4_TRIG_VAL.
				10 = Combination Mode: Alarm is initially triggered as described for Absolute Mode; the alarm then operates as described for Relative Mode.
				11 = Reserved
				Channel 4 must be disabled (ALM1_CH4_STS = 0) when updating this field.
R303170 (0x4A042)	15	ALM1_CH4_UPD	0	Channel 4 update control—Write 1 to indicate a new trigger value or pulse duration
ALM1_CTRL4				is ready to be applied.
				If Channel 4 is enabled and ALM1_CH4_UPD is set, the ALM1_CH4_TRIG_VAL and ALM1_CH4_PULSE_DUR settings are updated when the alarm is next triggered or by writing 1 to ALM1_CH4_START.
				If Channel 4 is disabled, the ALM1_CH4_UPD bit has no effect, and the ALM1_ CH4_TRIG_VAL and ALM1_CH4_PULSE_DUR settings are updated immediately when writing to the respective fields.
	4	ALM1 CH4		Channel 4 stop control—Write 1 to disable Channel 4
	-	STOP		
	0	ALM1_CH4_ START		Channel 4 start control—Write 1 to enable or restart Channel 4
R303172 (0x4A044)	31:0	ALM1_CH4_	0x0000	Channel 4 alarm trigger value
ALM1_TRIG_VAL4		TRIG_VAL[31:0]	_0000	
 R303174 (0x4A046)	31:0	ALM1_CH4_		Channel 4 alarm output pulse duration
ALM1_PULSE_DUR4		PULSĒ DUR[31:0]	_0000	The pulse duration is referenced to the count rate of the selected timer source
R303176 (0x4A048)	0	ALM1_CH4_STS	0	Channel 4 status
ALM1_STATUS4				0 = Disabled
				1 = Enabled

Table 4-39. Alarm (ALM1) Control (Cont.)

4.5.3 General-Purpose Timer

The CS42L92 incorporates a general-purpose timer, which supports a wide variety of uses. The general-purpose timer provides time-stamp data for the event logger; it also supports the watchdog and other miscellaneous time-based functions, providing additional capability for signal-processing applications.

4.5.3.1 Overview

The timer allows time-stamp information to be associated with external signal detection, and other system events, enabling real-time data to be more easily integrated into user applications. The timer allows many advanced functions to be implemented with a high degree of autonomy from a host processor.

The timer can use either internal system clocks, or external clock signals, as a reference. The selected reference is scaled down, using configurable dividers, to the required clock count frequency.

4.5.3.2 Timer Control

The reference clock for the timer is selected using TIMER1_REFCLK_SRC.



If SYSCLK, ASYNCCLK, or DSPCLK is selected, a lower clock frequency, derived from the applicable system clock, can be selected using the TIMER1_REFCLK_FREQ_SEL field (for SYSCLK or ASYNCCLK source) or the TIMER1_ DSPCLK_FREQ_SEL field (for DSPCLK source). The applicable division ratio is determined automatically, assuming the respective clock source has been correctly configured as described in Section 4.16.

Note that, depending on the DSPCLK frequency and the available clock dividers, the timer reference clock may differ from the selected clock if DSPCLK is the selected source. In most cases, the reference clock frequency equals or exceeds the requested frequency. A lower frequency is implemented if limited by either the DSPCLK frequency or the maximum TIMER1 clocking frequency.

If any source other than DSPCLK is selected, the clock can be further divided using TIMER1_REFCLK_DIV. Division ratios in the range 1 to 128 can be selected.

Note that, if DSPCLK is enabled, the CS42L92 synchronizes the selected reference clock to DSPCLK. As a result of this, if a non-DSPCLK is selected as source, the following additional constraints must be observed: the reference clock frequency (after TIMER1_REFCLK_FREQ_SEL and after TIMER1_REFCLK_DIV) must be less than DSPCLK / 3, and must be less than 12 MHz; it must also be close to 50% duty cycle. The TIMER1_REFCLK_DIV field can be used to ensure that these criteria are met.

One final division, controlled by TIMER1_PRESCALE, determines the timer count frequency. This field is valid for all clock reference sources; division ratios in the range 1 to 128 can be selected. The output from this division corresponds to the frequency at which the TIMER1_COUNT field is incremented (or decremented).

The maximum count value of the timer is determined by the TIMER1_MAX_COUNT field. This is the final count value (when counting up), or the initial count value (when counting down). The current value of the timer counter can be read from the TIMER1_CUR_COUNT field.

The timer is started by writing 1 to TIMER1_START. Note that, if the timer is already running, it restarts from its initial value. The timer is stopped by writing 1 to TIMER1_STOP. The count direction (up or down) is selected using the TIMER1_DIR bit.

The TIMER1_CONTINUOUS bit selects whether the timer automatically restarts after the end-of-count condition has been reached. The TIMER1_RUNNING_STS indicates whether the timer is running, or if it has stopped.

Note that the timer should be stopped before making any changes to the timer control registers. The timer configuration should only be changed if TIMER1_RUNNING_STS = 0.

4.5.3.3 Interrupts and GPIO Output

The timer status is an input to the interrupt control circuit and can be used to trigger an interrupt event after the final count value is reached; see Section 4.15. Note that the interrupt does not occur immediately when the final count value is reached; the interrupt is triggered at the point when the next update to the timer count value would be due.

The timer status can be output directly on a GPIO pin as an external indication of the timer activity. See Section 4.14 to configure a GPIO pin for this function.



4.5.3.4 Timer Block Diagram and Control Registers

The timer block is shown in Fig. 4-35.

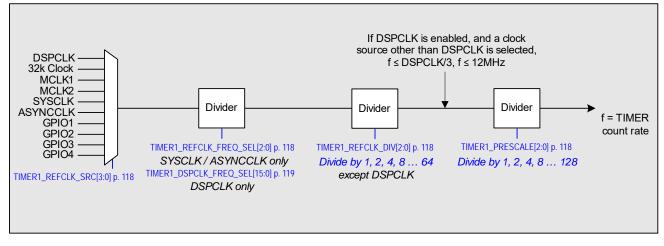


Figure 4-35. General-Purpose Timer

The timer control registers are described in Table 4-40.



Register Address	Bit	Label	Default	
R311296 (0x4_C000)	21	TIMER1_	0	Timer Continuous Mode select
Timer1_Control		CONTINUOUS		0 = Single mode
				1 = Continuous mode
				Timer must be stopped (TIMER1_RUNNING_STS = 0) when updating this field
	20	TIMER1_DIR	0	Timer Count Direction
				0 = Down
				1 = Up
				Timer must be stopped (TIMER1_RUNNING_STS = 0) when updating this field
	18:16	TIMER1_	000	Timer Count Rate Prescale
		PRESCALE[2:0]		000 = Divide by 1 011 = Divide by 8 110 = Divide by 64
				001 = Divide by 2 100 = Divide by 16 111 = Divide by 128
				010 = Divide by 4 101 = Divide by 32
				Timer must be stopped (TIMER1_RUNNING_STS = 0) when updating this field
	14:12	TIMER1_	000	Timer Reference Clock Divide (Not valid for DSPCLK source).
		REFCLK_DIV[2:0]		000 = Divide by 1 011 = Divide by 8 110 = Divide by 64
				001 = Divide by 2 100 = Divide by 16 111 = Divide by 128
				010 = Divide by 4 101 = Divide by 32
				If DSPCLK is enabled, and DSPCLK is not selected as source, the output frequency from this divider must be set less than or equal to DSPCLK / 3, and less than or equal to 12 MHz.
				If DSPCLK is disabled, the output of this divider is used as clock reference for the event logger. In this case, the divider output corresponds to the frequency of event logging opportunities on the respective modules.
				Timer must be stopped (TIMER1_RUNNING_STS = 0) when updating this field
	10:8	TIMER1	000	Timer Reference Frequency Select (SYSCLK or ASYNCCLK source)
		REFCLK_FREQ_		000 = 6.144 MHz (5.6448 MHz) 010 = 24.576 MHz (22.5792 MHz)
		SEL[2:0]		001 = 12.288 MHz (11.2896 MHz) 011 = 49.152 MHz (45.1584 MHz)
				All other codes are reserved.
				The selected frequency must be less than or equal to the frequency of the source.
				Timer must be stopped (TIMER1_RUNNING_STS = 0) when updating this field.
	3:0	TIMER1_	0000	Timer Reference Source Select.
		REFCLK_		Timer must be stopped (TIMER1_RUNNING_STS=0) when updating this field.
		SRC[3:0]		0000 = DSPCLK 1000 = SYSCLK 1110 = GPIO3
				0001 = 32-kHz clock 1001 = ASYNCCLK 1111 = GPIO4
				0100 = MCLK1 1100 = GPIO1 All other codes are
				0101 = MCLK2 1101 = GPIO2 reserved.
R311298 (0x4_C002)	31:0	TIMER1_MAX_		Timer Maximum Count.
Timer1_Count_Preset		COUNT[31:0]	_0000	Final count value (when counting up). Starting count value (when counting down). Timer must be stopped (TIMER1 RUNNING STS = 0) when updating this field.
R311302 (0x4_C006)	4	TIMER1 STOP	0	Timer Stop Control
Timer1_Start_and_	- T			Write 1 to stop.
Stop	0	TIMER1_START	0	Timer Start Control
				Write 1 to start.
				If the timer is already running, it restarts from its initial value.
R311304 (0x4 C008)	0	TIMER1	0	Timer Running Status
Timer1_Status		RUNNING_STS	0	0 = Timer stopped
				1 = Timer running

Table 4-40. General-Purpose Timer (TIMER1) Control



Register Address	Bit	Label	Default	Description		
R311306 (0x4_C00A)	31:0	TIMER1_CUR_	0x0000	Timer Current Count value		
Timer1_Count_ Readback		COUNT[31:0]				
R311308 (0x4_C00C)	15:0	TIMER1_	0x0000	Timer Reference Frequency Select (DSPCLK source)		
Timer1_DSP_Clock_		DSPCLK_FREQ_ SEL[15:0]		Coded as LSB = 1/64 MHz, Valid from 5.6 MHz to 148 MHz.		
Config			SEL[15:0]	SEL[15:0]	SEL[15:0]	
				Timer must be stopped (TIMER1_RUNNING_STS=0) when updating this field.		
R311310 (0x4_C00E)	15:0	TIMER1_	0x0000	Timer Reference Frequency (Read only)		
Timer1_DSP_Clock_	imer1_DSP_Clock_	DSPCLK_FREQ_		Only valid when DSPCLK is the selected clock source.		
Status	STS[15:0]			Coded as LSB = 1/64 MHz.		

Table 4-40. General-Purpose Timer (TIMER1) Control (Cont.)

4.5.4 DSP GPIO

The DSP GPIO function provides an advanced I/O capability, supporting enhanced flexibility for signal-processing applications.

4.5.4.1 Overview

The CS42L92 supports up to 16 GPIO pins, which can be assigned to application-specific functions. There are 2 dedicated GPIO pins; the remaining 14 GPIOs are implemented as alternate functions to a pin-specific capability.

The GPIOs can be used to provide status outputs and control signals to external hardware; the supported functions include interrupt output, FLL clock output, accessory detection status, and S/PDIF or PWM-coded audio channels; see Section 4.14.

The GPIOs can support miscellaneous logic input and output, interfacing directly with the integrated DSPs, or with the host application software. A basic level of I/O functionality is described in Section 4.14, under the configuration where $GPn_FN = 0x001$. The GPn_FN field selects the functionality for the respective pin, GPIOn.

The DSP GPIO pins are accessed using maskable sets of I/O control registers; this allows the selected combinations of GPIOs to be controlled with ease, regardless of how the allocation of GPIO pins has been implemented in hardware. In a typical use case, one GPIO mask is defined for each DSP function; this provides a highly efficient mechanism for the DSP to independently access the respective input and output signals.

4.5.4.2 DSP GPIO Control

The DSP GPIO function is selected by setting $GPn_FN = 0x002$ for the respective GPIO pin (where *n* identifies the applicable GPIO*n* pin).

Each DSP GPIO is controlled using bits that determine the direction (input/output) and the logic state (0/1) of the pin. These bits are replicated in eight control sets; each which can determine the logic level of any DSP GPIO.

Mask bits are provided within each control set, to determine which of the control sets has control of each DSP GPIO. To avoid logic contention, a DSP GPIO output must be controlled (unmasked) in a maximum of one control set at any time.

Note that write access to the direction control bits (DSPGPn_SETx_DIR) and level control bits (DSPGPn_SETx_LVL) is only valid when the channel (DSPGPn) is unmasked in the respective control set. Writes to these fields are implemented for the unmasked DSP GPIOs, and are ignored in respect of the masked DSP GPIOs. Note that the level control bits (DSPGPn_SETx_LVL) provide output level control only—they cannot be used to read the status of DSP GPIO inputs.

The logic level of the unmasked DSP GPIO outputs in any control set can be configured using a single register write. Writing to the output level control registers determines the logic level of the unmasked DSP GPIOs in that set only; all other outputs are unaffected.



DSP GPIO status bits are provided, indicating the logic level of every input or output pin that is configured as a DSP GPIO. The DSPGP n_STS bits also provide logic-level indication for any pin that is configured as a GPIO input, with GP $n_FN = 0x001$.Note that there is only one set of DSP GPIO status bits.

The status bits indicate the logic level of the DSP GPIO outputs. The respective pins are driven as outputs if configured as a DSP GPIO output, and unmasked in one of the control sets. Note that a DSP GPIO continues to be driven as an output, even if the mask bit is subsequently asserted in that set. The pin only ceases to be driven if it is configured as a DSP GPIO input and is unmasked in one of the control sets, or if the pin is configured as an input under a different GP*n*_FN field selection.

4.5.4.3 Common Functions to Standard GPIOs

The DSP GPIO functions are implemented alongside the standard GPIO capability, providing an alternative method of maskable I/O control for all of the GPIO pins. The DSP GPIO control bits in the register map are implemented in a manner that supports efficient read/write access for multiple GPIOs at once.

The DSP GPIO logic is shown in Fig. 4-36, which also shows the control fields relating to the standard GPIO.

The DSP GPIO function is selected by setting $GPn_FN = 0x002$ for the respective GPIO pin. Integrated pull-up and pull-down resistors are provided on each GPIO pin, which are also valid for DSP GPIO function. A bus keeper function is supported on the GPIO pins; this is enabled using the respective pull-up and pull-down control bits. The bus keeper function holds the logic level unchanged whenever the pin is undriven (e.g., if the signal is tristated). See Table 4-93 for details of the GPIO pull-up and pull-down control bits.



4.5.4.4 DSP GPIO Block Diagram and Control Registers

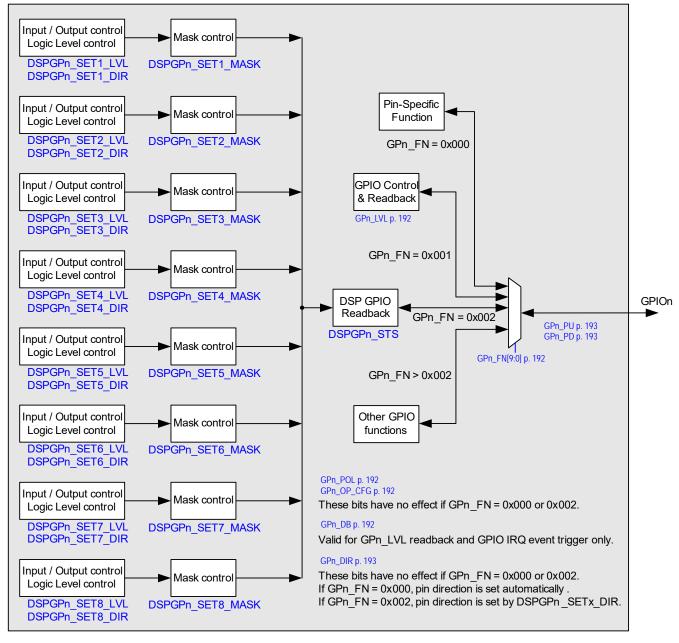


Figure 4-36. DSP GPIO Control

The control registers associated with the DSP GPIO are described in Table 4-41.



Register Address	Bit	Label	Default	Description
R315392 (0x4_D000)	15	DSPGP16_STS	0	DSPGP16 Status
DSPGP_Status_1				Valid for DSPGP input and output
	14	DSPGP15 STS	0	DSPGP15 Status
	13	DSPGP14 STS	0	DSPGP14 Status
	12	DSPGP13 STS	0	DSPGP13 Status
		DSPGP12 STS	0	DSPGP12 Status
		DSPGP11 STS	0	DSPGP11 Status
	9	DSPGP10 STS	0	DSPGP10 Status
	8	DSPGP9 STS	0	DSPGP9 Status
	7	DSPGP8 STS	0	DSPGP8 Status
	6	DSPGP7 STS	0	DSPGP7 Status
	5	DSPGP6 STS	0	DSPGP6 Status
	4	DSPGP5 STS	0	DSPGP5 Status
	3	DSPGP4_STS	0	DSPGP4 Status
	2	DSPGP3 STS	0	DSPGP3 Status
	1	DSPGP2 STS	0	DSPGP2 Status
	0	DSPGP1 STS	0	DSPGP1 Status
R315424 (0x4 D020)	-	DSPGP16 SETn MASK	1	DSP SETn GPIO16 Mask Control
DSPGP SET1 Mask 1	10		•	0 = Unmasked, 1 = Masked
R315456 (0x4 D040)				A GPIO pin should be unmasked in a maximum of one SET at any time.
DSPGP_SET2_Mask_1	14	DSPGP15_SETn_MASK	1	DSP SETn GPIO15 Mask Control
R315488 (0x4_D060)		DSPGP14 SETn MASK	1	DSP SETn GPIO14 Mask Control
DSPGP_SET3_Mask_1		DSPGP13_SETn_MASK	1	DSP SETn GPIO13 Mask Control
R315520 (0x4_D080)	11	DSPGP12_SETn_MASK	1	DSP SETn GPIO12 Mask Control
DSPGP_SET4_Mask_1	10	DSPGP11 SETn MASK	1	DSP SETn GPIO11 Mask Control
R315552 (0x4_D0A0)	9	DSPGP10_SETn_MASK	1	DSP SETn GPI010 Mask Control
DSPGP_SET5_Mask_1	8	DSPGP9_SETn_MASK	1	DSP SETn GPIO9 Mask Control
R315584 (0x4_D0C0) DSPGP_SET6_Mask_1	7	DSPGP8 SETn MASK	1	DSP SETn GPIO8 Mask Control
R315616 (0x4 D0E0)	6	DSPGP7 SETn MASK	1	DSP SETN GPI00 Mask Control
DSPGP_SET7_Mask_1	5	DSPGP6 SETn MASK	1	DSP SETn GPIO/ Mask Control
R315648 (0x4 D100)	4	DSPGP5 SETn MASK	1	DSP SETN GPI05 Mask Control
DSPGP_SET8_Mask_1	4	DSPGP4 SETn MASK	1	DSP SETN GPIOS Mask Control
	2	DSPGP3 SETn MASK	1	DSP SETN GPIO4 Mask Control
	2	DSPGP2 SETn MASK	1	DSP SETN GPIOS Mask Control
	0		1	DSP SETN GPI02 Mask Control
B215422 (0x4 D028)	-	DSPGP1_SETn_MASK DSPGP16 SETn DIR	1	DSP SETI GPI01 Mask Control
R315432 (0x4_D028) DSPGP SET1 Direction 1	15	DSPGP10_SETILDIR	I	0 = Output, 1 = Input
	14	DSPGP15_SETn_DIR	1	DSP SETn GPI015 Direction Control
DSPGP SET2 Direction 1			1	
R315496 (0x4 D068)		DSPGP14_SETn_DIR	1	DSP SETn GPIO14 Direction Control
DSPGP_SET3_Direction_1	-	DSPGP13_SETn_DIR	1	DSP SETn GPI013 Direction Control
R315528 (0x4_D088)	11	DSPGP12_SETn_DIR	1	DSP SETn GPI012 Direction Control
DSPGP_SET4_Direction_1	10	DSPGP11_SETn_DIR	1	DSP SETn GPIO11 Direction Control
R315560 (0x4_D0A8)	9	DSPGP10_SETn_DIR	1	DSP SETn GPI010 Direction Control
DSPGP_SET5_Direction_1	8	DSPGP9_SETn_DIR	1	DSP SETn GPIO9 Direction Control
R315592 (0x4_D0C8)	7	DSPGP8_SETn_DIR	1	DSP SETn GPIO8 Direction Control
DSPGP_SET6_Direction_1	6	DSPGP7_SETn_DIR	1	DSP SETn GPIO7 Direction Control
R315624 (0x4_D0E8) DSPGP_SET7_Direction_1	5	DSPGP6_SETn_DIR	1	DSP SETn GPIO6 Direction Control
R315656 (0x4 D108)	4	DSPGP5_SETn_DIR	1	DSP SETn GPIO5 Direction Control
DSPGP_SET8_Direction_1	3	DSPGP4_SETn_DIR	1	DSP SETn GPIO4 Direction Control
	2	DSPGP3_SETn_DIR	1	DSP SETn GPIO3 Direction Control
	1	DSPGP2_SETn_DIR	1	DSP SETn GPIO2 Direction Control
	0	DSPGP1_SETn_DIR	1	DSP SETn GPIO1 Direction Control

Table 4-41. DSP GPIO Control



Register Address	Bit	Label	Default	Description
R315440 (0x4_D030)	15	DSPGP16_SETn_LVL	0	DSP SETn GPIO16 Output Level
DSPGP_SET1_Level_1				0 = Logic 0, 1 = Logic 1
R315472 (0x4_D050)	14	DSPGP15_SETn_LVL	0	DSP SETn GPIO15 Output Level
DSPGP_SET2_Level_1	13	DSPGP14_SETn_LVL	0	DSP SETn GPIO14 Output Level
R315504 (0x4_D070) DSPGP SET3 Level 1	12	DSPGP13_SETn_LVL	0	DSP SETn GPIO13 Output Level
R315536 (0x4 D090)	11	DSPGP12_SETn_LVL	0	DSP SETn GPIO12 Output Level
DSPGP SET4 Level 1	10	DSPGP11_SETn_LVL	0	DSP SETn GPIO11 Output Level
R315568 (0x4 D0B0)	9	DSPGP10_SETn_LVL	0	DSP SETn GPIO10 Output Level
DSPGP_SET5_Level_1	8	DSPGP9_SETn_LVL	0	DSP SETn GPIO9 Output Level
R315600 (0x4_D0D0)	7	DSPGP8_SETn_LVL	0	DSP SETn GPIO8 Output Level
DSPGP_SET6_Level_1	6	DSPGP7_SETn_LVL	0	DSP SETn GPIO7 Output Level
R315632 (0x4_D0F0)	5	DSPGP6_SETn_LVL	0	DSP SETn GPIO6 Output Level
DSPGP_SET7_Level_1	4	DSPGP5_SETn_LVL	0	DSP SETn GPIO5 Output Level
R315664 (0x4_D110)	3	DSPGP4_SETn_LVL	0	DSP SETn GPIO4 Output Level
DSPGP_SET8_Level_1	2	DSPGP3_SETn_LVL	0	DSP SETn GPIO3 Output Level
	1	DSPGP2_SETn_LVL	0	DSP SETn GPIO2 Output Level
	0	DSPGP1_SETn_LVL	0	DSP SETn GPIO1 Output Level

Table 4-41. DSP GPIO Control (Cont.)

4.6 Digital Audio Interface

The CS42L92 provides three audio interfaces, AIF1–AIF3. Each interface is independently configurable on the respective transmit (TX) and receive (RX) paths. Each AIF supports up to eight channels of input and output signal paths.

The data sources for the audio interface transmit (TX) paths can be selected from any of the CS42L92 input signal paths, or from the digital-core processing functions. The audio interface receive (RX) paths can be selected as inputs to any of the digital-core processing functions or digital-core outputs. See Section 4.3 for details of the digital-core routing options.

The digital audio interfaces provide flexible connectivity for multiple processors and other audio devices. Typical connections include applications processor, baseband processor, and wireless transceiver. Note that the SLIMbus interface also provides digital audio input/output paths, providing options for additional interfaces. A typical configuration is shown in Fig. 4-37.



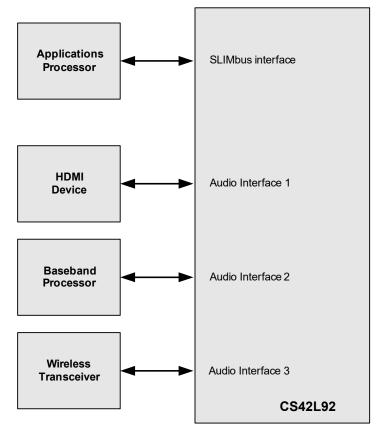


Figure 4-37. Typical AIF Connections

In the general case, the digital audio interface uses four pins:

- TXDAT: data output
- RXDAT: data input
- · BCLK: bit clock, for synchronization
- LRCLK: left/right data-alignment clock

In Master Mode, the clock signals BCLK and LRCLK are outputs from the CS42L92. In Slave Mode, these signals are inputs, as shown in Section 4.6.1.

The following interface formats are supported on AIF1-AIF3:

- DSP Mode A
- DSP Mode B
- I²S
- Left-justified

The left-justified and DSP-B formats are valid in Master Mode only (i.e., BCLK and LRCLK are outputs from the CS42L92). These modes cannot be supported in Slave Mode.

The audio interface formats are described in Section 4.6.2. The bit order is MSB-first in each case. Mono PCM operation can be supported using the DSP modes. Refer to Table 3-16 through Table 3-18 for signal timing information.

For typical applications, AIF data is encoded in 2's complement (signed, fixed-point) format. This format is compatible with all of the digital mixing and signal-processing functions on the CS42L92. Other data types, including floating point formats, can be supported using the DFCs. Note that, if unsigned or floating point data is present within the digital core, some restrictions on the valid signal routing options apply—see Section 4.3.13.



4.6.1 Master and Slave Mode Operation

The CS42L92 digital audio interfaces can operate as a master or slave, as shown in Fig. 4-38 and Fig. 4-39. The associated control bits are described in Section 4.7.

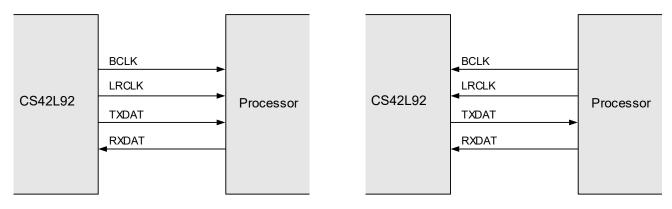


Figure 4-38. Master Mode

Figure 4-39. Slave Mode

4.6.2 Audio Data Formats

The CS42L92 digital audio interfaces can be configured to operate in I²S, left-justified, DSP-A, or DSP-B interface modes. Note that left-justified and DSP-B modes are valid in Master Mode only (i.e., BCLK and LRCLK are outputs from the CS42L92).

The digital audio interfaces also provide flexibility to support multiple slots of audio data within each LRCLK frame. This flexibility allows multiple audio channels to be supported within a single LRCLK frame.

The data formats described in this section are generic descriptions, assuming only one stereo pair of audio samples per LRCLK frame. In these cases, the AIF is configured to transmit (or receive) in the first available position in each frame (i.e., the Slot 0 position).

The options for multichannel operation are described in Section 4.6.3.

The audio data modes supported by the CS42L92 are described as follows. Note that the polarity of the BCLK and LRCLK signals can be inverted if required; the following descriptions all assume the default, noninverted polarity of these signals.

 In DSP modes, the left channel MSB is available on either the first (Mode B) or second (Mode A) rising edge of BCLK following a rising edge of LRCLK. Right-channel data immediately follows left channel data. Depending on word length, BCLK frequency, and sample rate, there may be unused BCLK cycles between the LSB of the right channel data and the next sample.

In Master Mode, the LRCLK output resembles the frame pulse shown in Fig. 4-40 and Fig. 4-41. In Slave Mode, it is possible to use any length of frame pulse less than 1/Fs, providing the falling edge of the frame pulse occurs at least one BCLK period before the rising edge of the next frame pulse.

PCM operation is supported in DSP interface mode. CS42L92 data that is output on the left channel is read as mono data by the receiving equipment. Mono PCM data received by the CS42L92 is treated as left-channel data. This may be routed to the left/right playback paths using the control fields described in Section 4.3.



DSP Mode A data format is shown in Fig. 4-40.

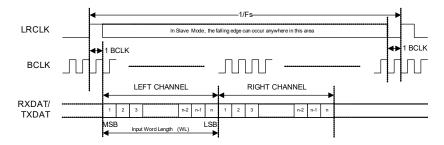


Figure 4-40. DSP Mode A Data Format

DSP Mode B data format is shown in Fig. 4-41.

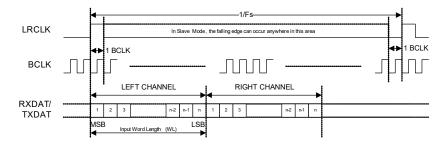


Figure 4-41. DSP Mode B Data Format

In I²S Mode, the MSB is available on the second rising edge of BCLK following a LRCLK transition. The other bits
up to the LSB are then transmitted in order. Depending on word length, BCLK frequency, and sample rate, there
may be unused BCLK cycles between the LSB of one sample and the MSB of the next.

I²S Mode data format is shown in Fig. 4-42.

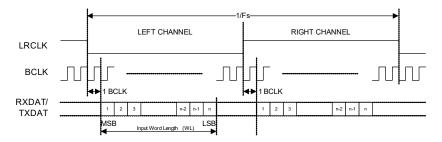


Figure 4-42. I²S Data Format (Assuming n-Bit Word Length)

 In Left-Justified Mode, the MSB is available on the first rising edge of BCLK following a LRCLK transition. The other bits up to the LSB are then transmitted in order. Depending on word length, BCLK frequency, and sample rate, there may be unused BCLK cycles before each LRCLK transition.

Left-Justified Mode data format is shown in Fig. 4-43.

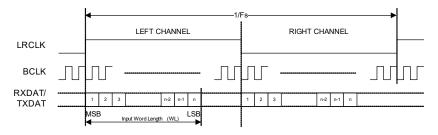


Figure 4-43. Left-Justified Data Format (Assuming n-Bit Word Length)



4.6.3 AIF Time-Slot Configuration

Multichannel operation is supported on AIF1–AIF3, with up to eight channels of input and output on each. A high degree of flexibility is provided to define the position of the audio samples within each LRCLK frame; the audio channel samples may be arranged in any order within the frame. Note that, on each interface, all input and output channels must operate at the same sample rate (Fs).

Each audio channel can be enabled or disabled independently on the transmit (TX) and receive (RX) signal paths. For each enabled channel, the audio samples are assigned to one time slot within the LRCLK frame.

In DSP modes, the time slots are ordered consecutively from the start of the LRCLK frame. In I²S and left-justified modes, the even-numbered time slots are arranged in the first half of the LRCLK frame, and the odd-numbered time slots are arranged in the second half of the frame.

The time slots are assigned independently for the transmit (TX) and receive (RX) signal paths. There is no requirement to assign every available time slot to an audio sample; slots may be left unused, if desired. Care is required, however, to ensure that no time slot is allocated to more than one audio channel.

The number of BCLK cycles within a slot is configurable; this is the slot length. The number of valid data bits within a slot is also configurable; this is the word length. The number of BCLK cycles per LRCLK frame must be configured; it must be ensured that there are enough BCLK cycles within each LRCLK frame to transmit or receive all of the enabled audio channels.

Examples of the AIF time-slot configurations are shown in Fig. 4-44 through Fig. 4-47. One example is shown for each of the four possible data formats.

Fig. 4-44 shows an example of DSP Mode A format. Four enabled audio channels are shown, allocated to time slots 0 through 3.

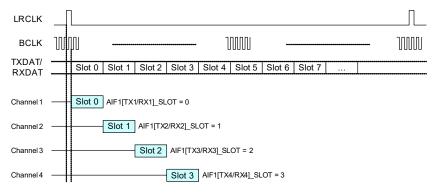


Figure 4-44. DSP Mode A Example



Fig. 4-45 shows an example of DSP Mode B format. Six enabled audio channels are shown, with time slots 4 and 5 unused.

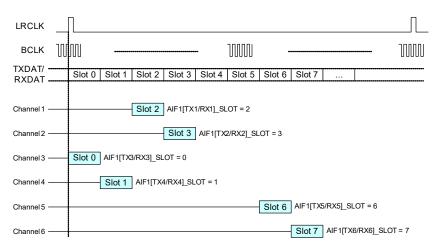


Figure 4-45. DSP Mode B Example

Fig. 4-46 shows an example of I²S format. Four enabled channels are shown, allocated to time slots 0 through 3.

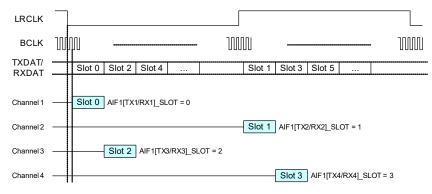


Figure 4-46. I²S Example

Fig. 4-47 shows an example of left-justified format. Six enabled channels are shown.

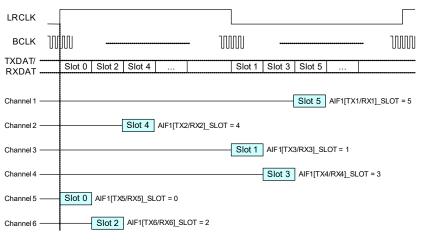


Figure 4-47. Left-Justified Example



4.6.4 TDM Operation Between Three or More Devices

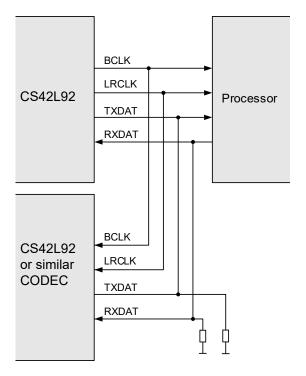
The AIF operation described in Section 4.6.3 illustrates how multiple audio channels can be interleaved on a single TXDAT or RXDAT pin. The interface uses TDM to allocate time periods to each audio channel in turn.

This form of TDM is implemented between two devices, using the electrical connections shown Fig. 4-38 or Fig. 4-39.

It is also possible to implement TDM between three or more devices. This allows one codec to receive audio data from two other devices simultaneously on a single audio interface, as shown in Fig. 4-48, Fig. 4-49, and Fig. 4-50.

The CS42L92 provides full support for TDM operation. The TXDAT pin can be tristated when not transmitting data, in order to allow other devices to transmit on the same wire. The behavior of the TXDAT pin is configurable, to allow maximum flexibility to interface with other devices in this way.

Typical configurations of TDM operation between three devices are shown in Fig. 4-48, Fig. 4-49, and Fig. 4-50.



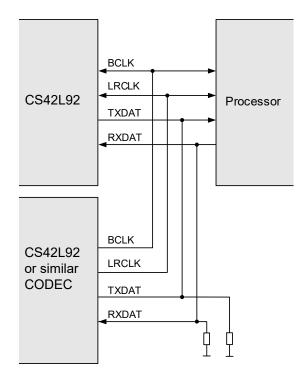


Figure 4-48. TDM with CS42L92 as Master

Figure 4-49. TDM with Other Codec as Master



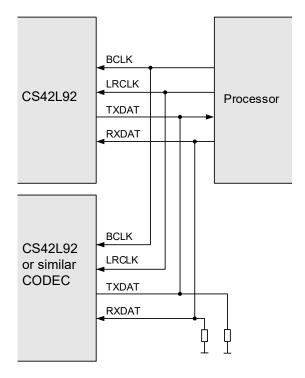


Figure 4-50. TDM with Processor as Master

4.7 Digital Audio Interface Control

This section describes the configuration of the CS42L92 digital audio interface paths.

Each AIF supports up to eight input signal paths and up to eight output signal paths. The digital audio interfaces can be configured as master or slave interfaces; mixed master/slave configurations are also possible.

Each input and output signal path can be independently enabled or disabled. The AIF output (TX) and AIF input (RX) paths use shared BCLK and LRCLK control signals.

The digital audio interface supports flexible data formats, selectable word length, configurable time-slot allocations, and TDM tristate control.

The AIF1 and AIF3 interfaces provide full support for 32-bit data words (input and output). Audio data samples up to 32 bits can be routed to the AIF1, AIF3, SLIMbus, S/PDIF, and DAC output paths. Note that other signal paths and signal-processing blocks within the digital core are limited to 24-bit data length; data samples are truncated to 24-bit length if they are routed through any function that does not support 32-bit data words.

The audio interfaces can be reconfigured while enabled, including changes to the LRCLK frame length and the channel time-slot configurations. Care is required to ensure that any on-the-fly reconfiguration does not cause corruption to the active signal paths. Wherever possible, it is recommended to disable all channels before changing the AIF configuration.

4.7.1 AIF Sample-Rate Control

The AIF RX inputs may be selected as input to the digital mixers or signal-processing functions within the CS42L92 digital core. The AIF TX outputs are derived from the respective output mixers.

The sample rate for each digital audio interface AIFn is configured using the respective AIFn_RATE field—see Table 4-26.

Note that sample-rate conversion is required when routing the AIF paths to any signal chain that is asynchronous or configured for a different sample rate.



4.7.2 AIF Pin Configuration

The external connections associated with each digital audio interface (AIF) are implemented on multifunction GPIO pins, which must be configured for the respective AIF functions when required. The AIF connections are pin-specific alternative functions available on specific GPIO pins. See Section 4.14 to configure the GPIO pins for AIF operation.

Integrated pull-up and pull-down resistors can be enabled on the AIF*n*LRCLK, AIF*n*BCLK and AIF*n*RXDAT pins. This is provided as part of the GPIO functionality, and provides a flexible capability for interfacing with other devices. Each pull-up and pull-down resistor can be configured independently using the fields described in Table 4-93.

If the pull-up and pull-down resistors are both enabled, the CS42L92 provides a bus keeper function on the respective pin. The bus-keeper function holds the logic level unchanged whenever the pin is undriven (e.g., if the signal is tristated).

4.7.3 AIF Master/Slave Control

The digital audio interfaces can operate in master or slave modes and also in mixed master/slave configurations. In Master Mode, the BCLK and LRCLK signals are generated by the CS42L92 when any of the respective digital audio interface channels is enabled. In Slave Mode, these outputs are disabled by default to allow another device to drive these pins.

Master Mode is selected on the AIF*n*BCLK pin by setting AIF*n*_BCLK_MSTR. In Master Mode, the AIF*n*BCLK signal is generated by the CS42L92 when one or more AIF*n* channels is enabled.

If the AIF*n*_BCLK_FRC bit is set in BCLK Master Mode, the AIF*n*BCLK signal is output at all times, including when none of the AIF*n* channels is enabled.

The AIF nBCLK signal can be inverted in master or slave modes using the AIF n_BCLK_INV bit.

Master Mode is selected on the AIF*n*LRCLK pin by setting AIF*n*_LRCLK_MSTR. In Master Mode, the AIF*n*LRCLK signal is generated by the CS42L92 when one or more AIF*n* channels is enabled.

If AIF*n*_LRCLK_FRC is set in LRCLK Master Mode, the AIF*n*LRCLK signal is output at all times, including when none of the AIF*n* channels is enabled. Note that AIF*n*LRCLK is derived from AIF*n*BCLK, and an internal or external AIF*n*BCLK signal must be present to generate AIF*n*LRCLK.

The AIF*n*LRCLK signal can be inverted in master or slave modes using the AIF*n*_LRCLK_INV bit.

The timing of the AIF*n*LRCLK signal is selectable using AIF*n*_LRCLK_ADV. If this bit is set, the LRCLK signal transition is advanced to the previous BCLK phase (as compared with the default behavior). Further details of this option, and conditions for valid use cases, are described in Section 4.7.3.1.

The AIF1 master/slave control registers are described in Table 4-42.

Table 4-42. AIF1 Mas	ter/Slave Control
----------------------	-------------------

Register Address	Bit	Label	Default	Description
R1280 (0x0500)	7	AIF1_	0	AIF1 Audio Interface BCLK Invert
AIF1_BCLK_Ctrl		BCLK_INV		0 = AIF1BCLK not inverted
				1 = AIF1BCLK inverted
	6	AIF1_	0	AIF1 Audio Interface BCLK Output Control
		BCLK_FRC		0 = Normal
				1 = AIF1BCLK always enabled in Master Mode
	5	AIF1_	0	AIF1 Audio Interface BCLK Master Select
		BCLK_		0 = AIF1BCLK Slave Mode
		MSTR		1 = AIF1BCLK Master Mode



Register Address	Bit	Label	Default	Description		
R1282 (0x0502)	4	AIF1_	0	AIF1 Audio Interface LRCLK Advance		
AIF1_Rx_Pin_Ctrl		LRCLK_		0 = Normal		
		ADV		1 = AIF1LRCLK transition is advanced to the previous BCLK phase		
	2	AIF1_	0	AIF1 Audio Interface LRCLK Invert		
		LRCLK_INV		0 = AIF1LRCLK not inverted		
				1 = AIF1LRCLK inverted		
	1	AIF1_	0	AIF1 Audio Interface LRCLK Output Control		
		LRCLK_		0 = Normal		
		FRC		1 = AIF1LRCLK always enabled in Master Mode		
	0	AIF1_	0	AIF1 Audio Interface LRCLK Master Select		
		LRCLK_		0 = AIF1LRCLK Slave Mode		
		MSTR		1 = AIF1LRCLK Master Mode		

Table 4-42. AIF1 Master/Slave Control (Cont.)

The AIF2 master/slave control registers are described in Table 4-43.

Table 4-43. AIF2 Master/Slave Contro	Table 4-43.
--------------------------------------	-------------

Register Address	Bit	Label	Default	Description		
R1344 (0x0540)	7	AIF2_BCLK_	0	AIF2 Audio Interface BCLK Invert		
AIF2_BCLK_Ctrl		INV		0 = AIF2BCLK not inverted		
				1 = AIF2BCLK inverted		
		AIF2_BCLK_	0	AIF2 Audio Interface BCLK Output Control		
		FRC		0 = Normal		
				1 = AIF2BCLK always enabled in Master Mode		
	5	AIF2_BCLK_	0	AIF2 Audio Interface BCLK Master Select		
		MSTR		0 = AIF2BCLK Slave Mode		
				1 = AIF2BCLK Master Mode		
R1346 (0x0542)			•	AIF2 Audio Interface LRCLK Advance		
AIF2_Rx_Pin_Ctrl				0 = Normal		
				1 = AIF2LRCLK transition is advanced to the previous BCLK phase		
	2	····		AIF2 Audio Interface LRCLK Invert		
	LRCLK_INV			0 = AIF2LRCLK not inverted		
				1 = AIF2LRCLK inverted		
	1 AIF20 LRCLK_FRC		0	AIF2 Audio Interface LRCLK Output Control		
				0 = Normal		
				1 = AIF2LRCLK always enabled in Master Mode		
	0	AIF2_	0	AIF2 Audio Interface LRCLK Master Select		
		LRCLK_		0 = AIF2LRCLK Slave Mode		
		MSTR		1 = AIF2LRCLK Master Mode		

The AIF3 master/slave control registers are described in Table 4-44.

Register Address	Bit	Label	Default	Description
R1408 (0x0580)	7	AIF3_BCLK_	0	AIF3 Audio Interface BCLK Invert
AIF3_BCLK_Ctrl		INV		0 = AIF3BCLK not inverted
				1 = AIF3BCLK inverted
	6	AIF3_BCLK_	0	AIF3 Audio Interface BCLK Output Control
		FRC		0 = Normal
				1 = AIF3BCLK always enabled in Master Mode
	5	AIF3_BCLK_	0	AIF3 Audio Interface BCLK Master Select
		MSTR		0 = AIF3BCLK Slave Mode
				1 = AIF3BCLK Master Mode



Register Address	Bit	Label	Default	Description		
R1410 (0x0582)	4	AIF3_	-	AIF3 Audio Interface LRCLK Advance		
AIF3_Rx_Pin_Ctrl		LRCLK_ADV		0 = Normal		
				1 = AIF3LRCLK transition is advanced to the previous BCLK phase		
	2	AIF3_	0	AIF3 Audio Interface LRCLK Invert		
		LRCLK_INV		0 = AIF3LRCLK not inverted		
				1 = AIF3LRCLK inverted		
	1	AIF3_		AIF3 Audio Interface LRCLK Output Control		
		LRCLK_FRC		0 = Normal		
				1 = AIF3LRCLK always enabled in Master Mode		
	0	AIF3_	0	AIF3 Audio Interface LRCLK Master Select		
		LRCLK_		0 = AIF3LRCLK Slave Mode		
		MSTR		1 = AIF3LRCLK Master Mode		

Table 4-44. AIF3 Master/Slave Control (Cont.)

4.7.3.1 LRCLK Advance

The timing of the AIF*n*LRCLK signal can be adjusted using AIF*n*_LRCLK_ADV. If this bit is set, the LRCLK signal transition is advanced to the previous BCLK phase (as compared with the default behavior).

The LRCLK-advance option (AIF*n*_LRCLK_ADV = 1) is valid for DSP-A mode only, operating in Master Mode.

Note: BCLK inversion must be enabled (AIF*n*_BCLK_INV = 1) if the LRCLK-advance option is enabled.

The adjusted interface timing (AIF $n_LRCLK_ADV = 1$), is shown in Fig. 4-51. The left-channel MSB is available on the second rising edge of BCLK, 1.5 BCLK cycles after the LRCLK rising edge—assuming the BCLK output is inverted.

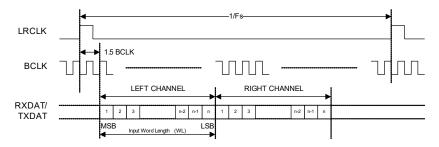


Figure 4-51. LRCLK advance—DSP-A Master Mode

4.7.4 AIF Signal Path Enable

The AIF1–AIF3 interfaces support up to eight input (RX) channels and up to eight output (TX) channels. Each channel is enabled or disabled using the bits defined in Table 4-45, Table 4-46 and Table 4-47.

The system clock, SYSCLK, must be configured and enabled before any audio path is enabled. The ASYNCCLK may also be required, depending on the path configuration. See Section 4.16 for details of the system clocks.

The audio interfaces can be reconfigured if enabled, including changes to the LRCLK frame length and the channel time-slot configurations. Care is required to ensure that this on-the-fly reconfiguration does not cause corruption to the active signal paths. Wherever possible, it is recommended to disable all channels before changing the AIF configuration.

The CS42L92 performs automatic checks to confirm that the SYSCLK and ASYNCCLK frequencies are high enough to support the commanded signal paths and processing functions. If the frequency is too low, an attempt to enable an AIF signal path fails. Note that active signal paths are not affected under such circumstances.



The AIF1 signal-path-enable bits are described in Table 4-45.

Table 4-45. AIF1 Signal Path Enable

Register Address	Bit	Label	Default	Description
R1305 (0x0519)	7	AIF1TX8_ENA	0	AIF1 Audio Interface TX Channel n Enable
AIF1_Tx_Enables	6	AIF1TX7_ENA	0	0 = Disabled
	5	AIF1TX6_ENA	0	1 = Enabled
	4	AIF1TX5_ENA	0	
	3	AIF1TX4_ENA	0	
	2	AIF1TX3_ENA	0	
	1	AIF1TX2_ENA	0	
	0	AIF1TX1_ENA	0	
R1306 (0x051A)	7	AIF1RX8_ENA	0	AIF1 Audio Interface RX Channel n Enable
AIF1_Rx_Enables	6	AIF1RX7_ENA	0	0 = Disabled
	5	AIF1RX6_ENA	0	1 = Enabled
	4	AIF1RX5_ENA	0	
	3	AIF1RX4_ENA	0	
	2	AIF1RX3_ENA	0	
	1	AIF1RX2_ENA	0	
	0	AIF1RX1_ENA	0	

The AIF2 signal-path-enable bits are described in Table 4-46.

Table 4-46. AIF2 Signal Path Enable

Register Address	Bit	Label	Default	Description
R1369 (0x0559)	7	AIF2TX8_ENA	0	AIF2 Audio Interface TX Channel <i>n</i> Enable
AIF2_Tx_Enables	6	AIF2TX7_ENA	0	0 = Disabled
	5	AIF2TX6_ENA	0	1 = Enabled
	4	AIF2TX5_ENA	0	
	3	AIF2TX4_ENA	0	
	2	AIF2TX3_ENA	0	
	1	AIF2TX2_ENA	0	
	0	AIF2TX1_ENA	0	
R1370 (0x055A)	7	AIF2RX8_ENA	0	AIF2 Audio Interface RX Channel <i>n</i> Enable
AIF2_Rx_Enables	6	AIF2RX7_ENA	0	0 = Disabled
	5	AIF2RX6_ENA	0	1 = Enabled
	4	AIF2RX5_ENA	0	
	3	AIF2RX4_ENA	0	
	2	AIF2RX3_ENA	0	
	1	AIF2RX2_ENA	0	
	0	AIF2RX1_ENA	0	

The AIF3 signal-path-enable bits are described in Table 4-47.

Register Address	Bit	Label	Default	Description
R1433 (0x0599)	7	AIF3TX8_ENA	0	AIF3 Audio Interface TX Channel n Enable
AIF3_Tx_Enables	6	AIF3TX7_ENA	0	0 = Disabled
	5	AIF3TX6_ENA	0	1 = Enabled
	4	AIF3TX5_ENA	0	
	3	AIF3TX4_ENA	0	
	2	AIF3TX3_ENA	0	
	1	AIF3TX2_ENA	0	
	0	AIF3TX1_ENA	0	



Register Address	Bit	Label	Default	Description
R1434 (0x059A)	7	AIF3RX8_ENA	0	AIF3 Audio Interface RX Channel <i>n</i> Enable
AIF3_Rx_Enables	6	AIF3RX7_ENA	0	0 = Disabled
	5	AIF3RX6_ENA	0	1 = Enabled
	4	AIF3RX5_ENA	0	
	3	AIF3RX4_ENA	0	
	2	AIF3RX3_ENA	0	
	1	AIF3RX2_ENA	0	
	0	AIF3RX1_ENA	0	

Table 4-47. AIF3 Signal Path Enable (Cont.)

4.7.5 AIF BCLK and LRCLK Control

The AIF nBCLK frequency is selected using the AIF n BCLK FREQ field. For each setting of this field, the actual frequency depends on whether AIFn is configured for a 48-kHz-related sample rate, as described in Table 4-48 through Table 4-50.

- If AIFn RATE < 1000 (Table 4-26), AIFn is referenced to the SYSCLK clocking domain and the applicable frequency depends upon the SAMPLE RATE 1, SAMPLE RATE 2 or SAMPLE RATE 3 fields.
- If AIFn_RATE ≥ 1000, AIFn is referenced to the ASYNCCLK clocking domain and the applicable frequency depends upon the ASYNC SAMPLE RATE 1 or ASYNC SAMPLE RATE 2 fields.

The selected AIF nBCLK rate must be less than or equal to SYSCLK/2, or ASYNCCLK/2, as applicable. See Section 4.16 for details of SYSCLK and ASYNCCLK clock domains, and the associated control registers.

The AIF*n*LRCLK frequency is controlled relative to AIF*n*BCLK by the AIF*n* BCPF divider.

Note that the BCLK rate must be configured in master and slave modes, using the AIF BCLK FREQ fields. The LRCLK rates only require to be configured in Master Mode.

The AIF1 BCLK/LRCLK control fields are described in Table 4-48.

Register Address	Bit	Label	Default		Description	
R1280	4:0	AIF1_	0x0C	AIF1BCLK Rate. The AIF1B	CLK rate must be less than or equ	al to SYSCLK/2.
(0x0500)		BCLK_		0x00–0x01 = Reserved	0x07 = 384 kHz (352.8 kHz)	0x0D = 3.072 MHz (2.8824 MHz)
AIF1_		FREQ[4:0]		0x02 = 64 kHz (58.8 kHz)	0x08 = 512 kHz (470.4 kHz)	0x0E = 4.096 MHz (3.7632 MHz)
BCLK_Ctrl				0x03 = 96 kHz (88.2 kHz)	0x09 = 768 kHz (705.6 kHz)	0x0F = 6.144 MHz (5.6448 MHz)
				0x04 = 128 kHz (117.6 kHz)	0x0A = 1.024 MHz (940.8 kHz)	0x10 = 8.192 MHz (7.5264 MHz)
				0x05 = 192 kHz (176.4 kHz)	0x0B = 1.536 MHz (1.4112 MHz)	0x11 = 12.288 MHz (11.2896 MHz)
				0x06 = 256 kHz (235.2 kHz)	0x0C = 2.048 MHz (1.8816 MHz)	0x12 = 24.576 MHz (22.5792 MHz)
				The frequencies in brackets	apply for 44.1 kHz–related sample	e rates only.
				/	is referenced to the SYSCLK clock s apply if SAMPLE_RATE_n = 012	·
				_ /	s referenced to the ASYNCCLK cl s apply if ASYNC_SAMPLE_RATE	

Table 4-48. AIF1 BCLK and LRCLK Control

ne e, the R1286 0x0040 AIF1LRCLK Rate. Selects the number of BCLK cycles per AIF1LRCLK frame. AIF1LRCLK clock = 12:0 AIF1 (0x0506) BCPF[12:0] AIF1BCLK/AIF1 BCPF. Integer (LSB = 1), Valid from 8 to 8191. AIF1 Rx BCLK Rate



The AIF2 BCLK/LRCLK control fields are described in Table 4-49.

Table 4-49.	AIF2 BCLK and	LRCLK Control
-------------	---------------	---------------

Register Address	Bit	Label	Default		Description	
R1344 (0x0540) AIF2_ BCLK_Ctrl		AIF2_ BCLK_ FREQ[4:0]	0x0C	0x00–0x01 = Reserved 0x02 = 64 kHz (58.8 kHz) 0x03 = 96 kHz (88.2 kHz) 0x04 = 128 kHz (117.6 kHz)	CLK rate must be less than or equ 0x07 = 384 kHz (352.8 kHz) 0x08 = 512 kHz (470.4 kHz) 0x09 = 768 kHz (705.6 kHz) 0x0A = 1.024 MHz (940.8 kHz)	0x0D = 3.072 MHz (2.8824 MHz) 0x0E = 4.096 MHz (3.7632 MHz) 0x0F = 6.144 MHz (5.6448 MHz) 0x10 = 8.192 MHz (7.5264 MHz)
				If AIF2_RATE < 1000, AIF2 i 44.1 kHz–related frequencie If AIF2_RATE \geq 1000, AIF2 i	N /	0x12 = 24.576 MHz (22.5792 MHz) e rates only. < domain. In this case, the XXX. lock domain. In this case, the
R1350 (0x0546) AIF2_Rx_ BCLK_Rate	12:0	AIF2_ BCPF[12:0]	0x0040	AIF2LRCLK Rate. Selects th AIF2BCLK/AIF2_BCPF. Integer (LSB = 1), Valid from		2LRCLK frame. AIF2LRCLK clock =

The AIF3 BCLK/LRCLK control fields are described in Table 4-50.

Table 4-50. AIF3 BCLK and LRCLK Control	
	_

Register Address	Bit	Label	Default		Description	
R1408 (0x0580) AIF3_ BCLK_Ctrl		AIF3_ BCLK_ FREQ[4:0]	0x0C	0x00-0x01 = Reserved 0x02 = 64 kHz (58.8 kHz) 0x03 = 96 kHz (88.2 kHz) 0x04 = 128 kHz (117.6 kHz) 0x05 = 192 kHz (176.4 kHz) 0x06 = 256 kHz (235.2 kHz) The frequencies in brackets at If AIF3_RATE < 1000, AIF3 is 44.1 kHz-related frequencies If AIF3_RATE ≥ 1000 , AIF3 is	CLK rate must be less than or equ 0x07 = 384 kHz (352.8 kHz) 0x08 = 512 kHz (470.4 kHz) 0x09 = 768 kHz (705.6 kHz) 0x0A = 1.024 MHz (940.8 kHz) 0x0B = 1.536 MHz (1.4112 MHz) 0x0C = 2.048 MHz (1.8816 MHz) apply for 44.1 kHz–related sample s referenced to the SYSCLK clock apply if SAMPLE_RATE_n = 012 s referenced to the ASYNCCLK cl apply if ASYNC SAMPLE RATE	0x0D = 3.072 MHz (2.8824 MHz) 0x0E = 4.096 MHz (3.7632 MHz) 0x0F = 6.144 MHz (5.6448 MHz) 0x10 = 8.192 MHz (7.5264 MHz) 0x11 = 12.288 MHz (11.2896 MHz) 0x12 = 24.576 MHz (22.5792 MHz) e rates only. c domain. In this case, the XX. ock domain. In this case, the
R1414 (0x0586) AIF3_Rx_ BCLK_Rate		AIF3_ BCPF[12:0]	0x0040	AIF3LRCLK Rate. Selects the AIF3BCLK/AIF3_BCPF. Integer (LSB = 1), Valid from	2	3LRCLK frame. AIF3LRCLK clock =

4.7.6 AIF Digital Audio Data Control

The fields controlling the audio data format, word length, and slot configurations for AIF1–AIF3 are described in Table 4-51 through Table 4-53 respectively.

Note that left-justified and DSP-B modes are valid in Master Mode only (i.e., BCLK and LRCLK are outputs from the CS42L92).

The AIF*n* slot length is the number of BCLK cycles in one time slot within the overall LRCLK frame. The word length is the number of valid data bits within each time slot. If the word length is less than the slot length, there are unused BCLK cycles at the end of each time slot. The AIF*n* word length and slot length are independently selectable for the input (RX) and output (TX) paths.

For each AIF input (RX) and AIF output (TX) channel, the position of the audio data sample within the LRCLK frame is configurable. The x_SLOT fields define the time-slot position of the audio sample for the associated audio channel. Valid selections are Slot 0 upwards. The time slots are numbered as shown in Fig. 4-44 through Fig. 4-47.



Note that, in DSP modes, the time slots are ordered consecutively from the start of the LRCLK frame. In I²S and left-justified modes, the even-numbered time slots are arranged in the first half of the LRCLK frame, and the odd-numbered time slots are arranged in the second half of the frame.

The AIF1 data control fields are described in Table 4-51.

Register Address	Bit	Label	Default	Description
R1284 (0x0504)	2:0	AIF1_FMT[2:0]	000	AIF1 Audio Interface Format
AIF1_Format				000 = DSP Mode A
				001 = DSP Mode B
				$010 = I^2S \mod$
				011 = Left-Justified mode
				Other codes are reserved.
R1287 (0x0507)	13:8	AIF1TX_WL[5:0]	0x18	AIF1 TX Word Length (Number of valid data bits per slot)
AIF1_Frame_Ctrl_1				Integer (LSB = 1); Valid from 16 to 32
	7:0	AIF1TX_SLOT_	0x18	AIF1 TX Slot Length (Number of BCLK cycles per slot)
		LEN[7:0]		Integer (LSB = 1); Valid from 16 to 128
R1288 (0x0508)	13:8	AIF1RX_WL[5:0]	0x18	AIF1 RX Word Length (Number of valid data bits per slot)
AIF1_Frame_Ctrl_2				Integer (LSB = 1); Valid from 16 to 32
	7:0	AIF1RX_SLOT_	0x18	AIF1 RX Slot Length (Number of BCLK cycles per slot)
		LEN[7:0]		Integer (LSB = 1); Valid from 16 to 128
R1289 (0x0509)	5:0	AIF1TX1_SLOT[5:0]	0x0	AIF1 TX Channel n Slot position
to	5:0	AIF1TX2_SLOT[5:0]	0x1	Defines the TX time slot position of the Channel n audio sample
R1296 (0x0510)	5:0	AIF1TX3_SLOT[5:0]	0x2	Integer (LSB=1); Valid from 0 to 63
	5:0	AIF1TX4_SLOT[5:0]	0x3	
	5:0	AIF1TX5_SLOT[5:0]	0x4	
	5:0	AIF1TX6_SLOT[5:0]	0x5	
	5:0	AIF1TX7_SLOT[5:0]	0x6	
	5:0	AIF1TX8_SLOT[5:0]	0x7	7
R1297 (0x0511)	5:0	AIF1RX1_SLOT[5:0]	0x0	AIF1 RX Channel n Slot position
to	5:0	AIF1RX2_SLOT[5:0]	0x1	Defines the RX time slot position of the Channel n audio sample
R1304 (0x0518)	5:0	AIF1RX3_SLOT[5:0]	0x2	Integer (LSB=1); Valid from 0 to 63
	5:0	AIF1RX4_SLOT[5:0]	0x3	7
	5:0	AIF1RX5_SLOT[5:0]	0x4	1
	5:0	AIF1RX6_SLOT[5:0]	0x5	7
	5:0	AIF1RX7_SLOT[5:0]	0x6	1
	5:0	AIF1RX8_SLOT[5:0]	0x7	1

Table 4-51. AIF1 Digital Audio Data Control

The AIF2 data control fields are described in Table 4-52.

Table 4-52.	AIF2 Digital Audio Data Control

Register Address	Bit	Label	Default	Description
R1348 (0x0544)	2:0	AIF2_FMT[2:0]	000	AIF2 Audio Interface Format
AIF2_Format				000 = DSP Mode A
				001 = DSP Mode B
				010 = I ² S mode
				011 = Left-Justified mode
				Other codes are reserved.
R1351 (0x0547)	13:8	AIF2TX_WL[5:0]	0x18	AIF2 TX Word Length
AIF2_Frame_Ctrl_1				(Number of valid data bits per slot)
				Integer (LSB = 1); Valid from 16 to 32
	7:0	AIF2TX_SLOT_	0x18	AIF2 TX Slot Length
		LEN[7:0]		(Number of BCLK cycles per slot)
				Integer (LSB = 1); Valid from 16 to 128



Register Address	Bit	Label	Default	Description
R1352 (0x0548)	13:8	AIF2RX_WL[5:0]	0x18	AIF2 RX Word Length
AIF2_Frame_Ctrl_2				(Number of valid data bits per slot)
				Integer (LSB = 1); Valid from 16 to 32
	7:0	AIF2RX_SLOT_	0x18	AIF2 RX Slot Length
		LEN[7:0]		(Number of BCLK cycles per slot)
				Integer (LSB = 1); Valid from 16 to 128
R1353 (0x0549)	5:0	AIF2TX1_SLOT[5:0]	0x0	AIF2 TX Channel n Slot position
to	5:0	AIF2TX2_SLOT[5:0]	0x1	Defines the TX time slot position of the Channel n audio sample
R1360 (0x0550)	5:0	AIF2TX3_SLOT[5:0]	0x2	Integer (LSB=1); Valid from 0 to 63
	5:0	AIF2TX4_SLOT[5:0]	0x3	
	5:0	AIF2TX5_SLOT[5:0]	0x4	
	5:0	AIF2TX6_SLOT[5:0]	0x5	
	5:0	AIF2TX7_SLOT[5:0]	0x6	
	5:0	AIF2TX8_SLOT[5:0]	0x7	
R1361 (0x0551)	5:0	AIF2RX1_SLOT[5:0]	0x0	AIF2 RX Channel n Slot position
to	5:0	AIF2RX2_SLOT[5:0]	0x1	Defines the RX time slot position of the Channel n audio sample
R1368 (0x0558)	5:0	AIF2RX3_SLOT[5:0]	0x2	Integer (LSB=1); Valid from 0 to 63
	5:0	AIF2RX4_SLOT[5:0]	0x3	
	5:0	AIF2RX5_SLOT[5:0]	0x4	
	5:0	AIF2RX6_SLOT[5:0]	0x5	
	5:0	AIF2RX7_SLOT[5:0]	0x6	
	5:0	AIF2RX8_SLOT[5:0]	0x7	

Table 4-52. AIF2 Digital Audio Data Control (Cont.)

The AIF3 data control fields are described in Table 4-53.

Table 4-53.	AIF3 Digital	Audio Data	Control
-------------	--------------	------------	---------

Register Address	Bit	Label	Default	Description
R1412 (0x0584)	2:0	AIF3_FMT[2:0]	000	AIF3 Audio Interface Format
AIF3_Format				000 = DSP Mode A
				001 = DSP Mode B
				010 = I ² S mode
				011 = Left-Justified mode
				Other codes are reserved.
R1415 (0x0587)	13:8	AIF3TX_WL[5:0]	0x18	AIF3 TX Word Length (Number of valid data bits per slot)
AIF3_Frame_Ctrl_1				Integer (LSB = 1); Valid from 16 to 32
	7:0	AIF3TX_SLOT_	0x18	AIF3 TX Slot Length (Number of BCLK cycles per slot)
		LEN[7:0]		Integer (LSB = 1); Valid from 16 to 128
R1416 (0x0588)	13:8	AIF3RX_WL[5:0]	0x18	AIF3 RX Word Length (Number of valid data bits per slot)
AIF3_Frame_Ctrl_2				Integer (LSB = 1); Valid from 16 to 32
	7:0	AIF3RX_SLOT_	0x18	AIF3 RX Slot Length (Number of BCLK cycles per slot)
		LEN[7:0]		Integer (LSB = 1); Valid from 16 to 128
R1417 (0x0589)	5:0	AIF3TX1_SLOT[5:0]	0x0	AIF3 TX Channel n Slot position
to	5:0	AIF3TX2_SLOT[5:0]	0x1	Defines the TX time slot position of the Channel n audio sample
R1424 (0x0590)	5:0	AIF3TX3_SLOT[5:0]	0x2	Integer (LSB=1); Valid from 0 to 63
	5:0	AIF3TX4_SLOT[5:0]	0x3	
	5:0	AIF3TX5_SLOT[5:0]	0x4	
	5:0	AIF3TX6_SLOT[5:0]	0x5	
	5:0	AIF3TX7_SLOT[5:0]	0x6	
	5:0	AIF3TX8_SLOT[5:0]	0x7	



Register Address	Bit	Label	Default	Description
R1425 (0x0591)	5:0	AIF3RX1_SLOT[5:0]	0x0	AIF3 RX Channel n Slot position
to	5:0	AIF3RX2_SLOT[5:0]	0x1	Defines the RX time slot position of the Channel n audio sample
R1432 (0x0598)	5:0	AIF3RX3_SLOT[5:0]	0x2	Integer (LSB=1); Valid from 0 to 63
	5:0	AIF3RX4_SLOT[5:0]	0x3	
	5:0	AIF3RX5_SLOT[5:0]	0x4	
	5:0	AIF3RX6_SLOT[5:0]	0x5	
	5:0	AIF3RX7_SLOT[5:0]	0x6	1
	5:0	AIF3RX8_SLOT[5:0]	0x7	

Table 4-53. AIF3 Digital Audio Data Control (Cont.)

4.7.7 AIF TDM and Tristate Control

The AIF*n* output pins are tristated when the AIF*n*_TRI bit is set. Note that this function only affects output pins configured for the respective AIF*n* function—a GPIO pin that is configured for a different function is not affected by AIF*n*_TRI. See Section 4.14 to configure the GPIO pins.

Under default conditions, the AIF*n*TXDAT output is held at Logic 0 when the CS42L92 is not transmitting data (i.e., during time slots that are not enabled for output by the CS42L92). If the AIF*n*TX_DAT_TRI bit is set, the CS42L92 tristates the respective AIF*n*TXDAT pin when not transmitting data, allowing other devices to drive the AIF*n*TXDAT connection.

The AIF1 TDM and tristate control fields are described in Table 4-54.

Table 4-54. AIF1 TDM and Tristate Control

Register Address	Bit	Label	Default	Description
R1281 (0x0501)	5	AIF1TX_DAT_TRI	0	AIF1TXDAT Tristate Control
AIF1_Tx_Pin_Ctrl				0 = Logic 0 during unused time slots
				1 = Tristated during unused time slots
R1283 (0x0503)	6	AIF1_TRI	0	AIF1 Audio Interface Tristate Control
AIF1_Rate_Ctrl				0 = Normal
				1 = AIF1 Outputs are tristated
				Note that this bit only affects output pins configured for the respective AIF1 function.

The AIF2 TDM and tristate control fields are described in Table 4-55.

Table 4-55. AIF2 TDM and Tristate Control

Register Address	Bit	Label	Default	Description
R1345 (0x0541)	5	AIF2TX_DAT_TRI	0	AIF2TXDAT Tristate Control
AIF2_Tx_Pin_Ctrl				0 = Logic 0 during unused time slots
				1 = Tristated during unused time slots
R1347 (0x0543)	6	AIF2_TRI	0	AIF2 Audio Interface Tristate Control
AIF2_Rate_Ctrl				0 = Normal
				1 = AIF2 Outputs are tristated
				Note that this bit only affects output pins configured for the respective AIF2 function.

The AIF3 TDM and tristate control fields are described in Table 4-56.

	Table 4-56.	AIF3 TDM a	nd Tristate	Control
--	-------------	------------	-------------	---------

Register Address	Bit	Label	Default	Description
R1409 (0x0581)	5	AIF3TX_DAT_TRI	0	AIF3TXDAT Tristate Control
AIF3_Tx_Pin_Ctrl				0 = Logic 0 during unused time slots
				1 = Tristated during unused time slots
R1411 (0x0583)	6	AIF3_TRI	0	AIF3 Audio Interface Tristate Control
AIF3_Rate_Ctrl				0 = Normal
				1 = AIF3 Outputs are tristated
				Note that this bit only affects output pins configured for the respective AIF3 function.



4.8 SLIMbus Interface

The SLIMbus protocol is highly configurable and adaptable, supporting multiple audio signal paths, and mixed sample rates simultaneously. It also supports control messaging and associated communications between devices.

4.8.1 SLIMbus Devices

The SLIMbus components comprise different device classes (manager, framer, interface, generic). Each component on the bus has an interface device, which provides bus management services for the respective component. One or more components on the bus provide manager and framer device functions; the manager has the capabilities to administer the bus, although the framer is responsible for driving the CLK line and for driving the DATA required to establish the frame structure on the bus. Note that only one manager and one framer device is active at any time. The framer function can be transferred between devices when required. Generic devices provide the basic SLIMbus functionality for the associated ports, and for the transport protocol by which audio signal paths are established on the bus.

4.8.2 SLIMbus Frame Structure

The SLIMbus bit stream is formatted within a defined structure of cells, slots, subframes, frames, and superframes:

- A single data bit is known as a cell.
- Four cells make a slot.
- A frame consists of 192 slots.
- Eight frames make a superframe.

The bit stream structure is configurable to some extent, but the superframe definition always comprises 1536 slots. The transmitted/received bit rate can be configured according to system requirements and can be changed dynamically without interruption to active audio paths.

The SLIMbus CLK frequency (also the bus bit rate) is defined by a root frequency (RF) and a clock gear (CG). In the top clock gear (Gear 10), the CLK frequency is equal to the root frequency. Each reduction in the clock gear halves the CLK frequency, and doubles the duration of the superframe.

The SLIMbus bandwidth typically comprises control space (for bus messages, synchronization, etc.) and data space (for audio paths). The precise allocation is configurable and can be entirely control space, if required.

The subframe definition comprises the number of slots per subframe (6, 8, 24 or 32) and the number of these slots per subframe allocated as control space. The applicable combination of subframe length and control space width are defined by the Subframe Mode (SM) parameter.

The SLIMbus frame always comprises 192 slots, regardless of the subframe definition. A number of slots are allocated to control space, as noted above; the remaining slots are allocated to data space. Some of the control space is required for framing information and for the guide channel (see Section 4.8.3); the remainder of the control space are allocated to the message channel.

Multiline SLIMbus comprises one or more secondary data line, supporting additional bandwidth and flexibility for data transfer over the bus. All data lines are synchronized to the bus clock; the RF and CG parameters are common to all data lines. Note that control space is allocated on the primary data line only—secondary lines are used exclusively for data space. Accordingly, the SM parameter is defined for the primary line only.

4.8.3 Control Space

Framing information is provided in slots 0 and 96 of every frame. Slot 0 contains a 4-bit synchronization code; slot 96 contains the 32-bit framing information, transmitted 4 bits at a time over the eight frames that make up the SLIMbus superframe. The clock gear, root frequency, subframe configuration, along with some other parameters, are encoded within the framing information.



The guide channel occupies two slots within Frame 0. This provides the necessary information for a SLIMbus component to acquire and verify the frame synchronization. The guide channel occupies the first two control space slots within the first frame of the bit stream, excluding the framing information slots. Note that the exact slot allocation depends upon the applicable subframe mode.

The message channel is allocated all of the control space not used by the framing information or the guide channel. The message channel enables SLIMbus devices to communicate with each other, using a priority-based mechanism defined in the MIPI specification.

Messages may be broadcast to all devices on the bus, or can be addressed to specific devices using their allocated logical address (LA) or enumeration address (EA). Note that, device-specific messages are directed to a particular device (i.e., manager, framer, interface, or generic) within a component on the bus.

4.8.4 Data Space

The data space can be organized into a maximum of 256 data channels. Each channel, identified by a unique channel number (CN), is a stream of one or more contiguous slots, organized in a consistent data structure that repeats at a fixed interval.

A data channel is defined by its segment length (SL), (number of contiguous slots allocated), segment interval (spacing between the first slots of successive segments), and segment offset (the slot number of the first allocated slot within the superframe). The segment interval and segment offset are collectively defined by a segment distribution (SD), by which the SLIMbus manager may configure or reconfigure any data channel.

Each segment may comprise TAG, AUX, and DATA portions. Any of these portions may have a length of zero; the exact composition depends on the transport protocol (TP) for the associated channel. The DATA portion must be wide enough to accommodate one full word of the data channel contents. Data words cannot be spread across multiple segments.

The segment interval for each data channel represents the minimum spacing between consecutive data samples for that channel. (Note that the minimum spacing applies if every allocated segment is populated with new data; in many cases, additional bandwidth is allocated and not every allocated segment is used.)

The segment interval gives rise to segment windows for each data channel, aligned to the start of every superframe. The segment window boundaries define the times within which each new data sample must be buffered, ready for transmission—adherence to these fixed boundaries allows slot allocations to be moved within a segment window, without altering the signal latency. The segment interval may be either shorter or longer than the frame length, but there is always an integer number of segment windows per superframe.

To transfer data between devices on the SLIMbus interface, a data channel connection is established between a source and one or more destination (sink) ports. A unique port number (PN) address is defined for every active TX/RX port. Multiple data channels can share the same port address by assigning different end point (EP) values to each channel.

The TP defines the flow control or handshaking method used by the ports associated with a data channel. The applicable flow control modes depend on the relationship between the audio sample rate (flow rate) and the SLIMbus CLK frequency. If the two rates are synchronized and integer related, no flow control is needed. In other cases, the flow may be regulated by the use of a presence bit, which can be set by the source device (pushed protocol) or by the sink device (pulled protocol).

The data-channel structure is defined in terms of the TP, SD, SL, and data line (LN) parameters. For multiline operation, the LN value identifies the data line on which the channel is present. Note that the mapping of secondary data lines (1–7) with respect to the secondary data pins is configurable on each multiline SLIMbus component, using value elements associated with the respective interface device. See Section 4.9.3 for details of value elements.

The data-channel content definition includes a presence rate (PR) parameter (describing the nominal sample rate for the audio channel) and a frequency locked (FL) bit (identifying whether the data source is synchronized to the SLIMbus CLK). The data length (DL) parameter defines the size of each data sample (number of slots). The auxiliary bits format (AF) and data type (DT) parameters provide support for non-PCM encoded data channels; the channel link (CL) parameter is an indicator that channel CN is related to the previous channel, CN-1.



For a given root frequency and clock gear, the SL and SD parameters define the amount of SLIMbus bandwidth that is allocated to a given data channel. The minimum bandwidth requirements of a data channel are represented by the presence rate (PR) and data length (DL) parameters. The allocated SLIMbus bandwidth must be equal to or greater than the bandwidth of the data to be transferred.

The segment interval defines the repetition rate of the SLIMbus slots allocated to consecutive data samples for a given data channel. The presence rate (PR) is the nominal sample rate of the audio path. The segment rate (determined by the segment interval value) must be equal to or greater than the presence rate for a given data channel. The following constraints must be observed when configuring a SLIMbus channel:

- If pushed or pulled transport protocol is selected, the segment rate must be greater than the presence rate to ensure that samples are not dropped as a result of clock drift.
- If isochronous transport protocol is selected, the segment rate must be equal to the presence rate. Isochronous
 transport protocol should be selected only if the data source is frequency locked to the SLIMbus CLK (i.e., the data
 source is synchronized to the SLIMbus framer device).

4.9 SLIMbus Control Sequences

This section describes the messages and general protocol associated with the SLIMbus system.

Note: The SLIMbus specification permits flexibility in core message support for different components. See Section 4.10 for details regarding which messages are supported on each of the SLIMbus devices present on the CS42L92.

4.9.1 Device Management and Configuration

This section describes the SLIMbus messages associated with configuring all devices on the SLIMbus interface.

When the SLIMbus interface starts up, it is required that only one component provides the manager and framer device functions. Other devices can request connection to the bus after they have gained synchronization.

The REPORT_PRESENT (DC, DCV) message may be issued by devices attempting to connect to the bus. The payload of this message contains the device class (DC) and device class version (DCV) parameters, describing the type of device that is attempting to connect. This message may be issued autonomously by the connecting device, or else in response to a REQUEST_SELF_ANNOUNCEMENT message from the manager device.

After positively acknowledging the REPORT_PRESENT message, the manager device then issues the ASSIGN_LOGICAL_ADDRESS (LA) message to allow the other device to connect to the bus. The payload of this message contains the logical address (LA) parameter only; this is the unique address by which the connected device sends and receives SLIMbus messages. The device is then said to be enumerated.

Once a device has been successfully connected to the bus, the logical address (LA) parameter can be changed at any time using the CHANGE_LOGICAL_ADDRESS (LA) message.

The RESET_DEVICE message commands an individual SLIMbus device to perform its reset procedure. As part of the reset, all associated ports are reset, and any associated data channels are canceled. Note that, if the RESET_DEVICE command is issued to an interface device, it causes a component reset (i.e., all devices within the associated component are reset). Under a component reset, every associated device releases its logical address, and the component becomes disconnected from the bus.

4.9.2 Information Management

A memory map of information elements is defined for each device. This is arranged in 3 x 1-kB blocks, comprising core value elements, device class-specific value elements, and user value elements respectively, as described in the MIPI specification. Note that the contents of the user information portion for each CS42L92 SLIMbus device are reserved.

Read/write access is implemented using the messages described as follows. Specific elements within the information map are identified using the element code (EC) parameter. In the case of read access, a unique transaction ID (TID) is assigned to each message relating to a particular read/write request.

- The REQUEST_INFORMATION (TID, EC) message is used to instruct a device to respond with the indicated information. The payload of this message contains the transaction ID (TID) and the element code (EC).
- The REQUEST_CLEAR_INFORMATION (TID, EC, CM) message is used to instruct a device to respond with the indicated information, and also to clear all, or parts, of the same information slice. The payload of this message contains the transaction ID (TID), element code (EC), and clear mask (CM). The clear mask field is used to select which elements are to be cleared as part of the instruction.
- The REPLY_INFORMATION (TID, IS) message is used to provide output of a requested parameter. The payload of this message contains the transaction ID (TID) and the information slice (IS). The information slice bytes contain the value of the requested parameter.
- The CLEAR_INFORMATION (EC, CM) message is used to clear all, or parts, of the indicated information slice. The payload of this message contains the element code (EC) and clear mask (CM). The clear mask field is used to select which elements are to be cleared as part of the instruction.
- The REPORT_INFORMATION (EC, IS) message is used to inform other devices about a change in a specified element in the information map. The payload of this message contains the element code (EC) and the information slice (IS). The information slice bytes contain the new value of the applicable parameter.

4.9.3 Value Management (Including Register Access)

A memory map of value elements is defined for each device. This is arranged in 3 x 1-kB blocks, comprising core value elements, device class-specific value elements, and user value elements respectively, as described in the MIPI specification. These elements are typically parameters used to configure device behavior.

The user value elements of the interface device are used on CS42L92 to support read/write access to the register map. Details of how to access specific registers are described in Section 4.10. Note that, with the exception of the user value elements of the interface device, the contents of the user value portion for each CS42L92 SLIMbus device are reserved.

Read/write access is implemented using the messages described as follows. Specific elements within the value map are identified using the element code (EC) parameter. In the case of read access, a unique transaction ID (TID) is assigned to each message relating to a particular read/write request.

- The REQUEST_VALUE (TID, EC) message is used to instruct a device to respond with the indicated information. The payload of this message contains the transaction ID (TID) and the element code (EC).
- The REPLY_VALUE (TID, VS) message is used to provide output of a requested parameter. The payload of this
 message contains the transaction ID (TID) and the value slice (VS). The value slice bytes contain the value of the
 requested parameter.
- The CHANGE_VALUE (EC, VU) message is used to write data to a specified element in the value map. The payload of this message contains the element code (EC) and the value update (VU). The value update bytes contain the new value of the applicable parameter.

4.9.4 Frame and Clocking Management

This section describes the SLIMbus messages associated with changing the frame or clocking configuration. One or more configuration messages may be issued as part of a reconfiguration sequence; all of the updated parameters become active at once, when the reconfiguration boundary is reached.

- The BEGIN_RECONFIGURATION message is issued to define a reconfiguration boundary point: subsequent NEXT_* messages become active at the first valid superframe boundary following receipt of the RECONFIGURE_ NOW message. (A valid boundary must be at least two slots after the end of the RECONFIGURE_NOW message.) Both of these messages have no payload content.
- The NEXT_ACTIVE_FRAMER (LAIF, NCo, NCi) message is used to select a new device as the active framer. The payload of this message includes the logical address, incoming framer (LAIF). Two other fields (NCo, NCi) define the number of clock cycles for which the CLK line shall be inactive during the handover.
- The NEXT_SUBFRAME_MODE (SM) and NEXT_CLOCK_GEAR (CG) messages are used to reconfigure the SLIMbus clocking or framing definition. The payload of each is the respective subframe mode (SM) or clock gear (CG) respectively.



- The NEXT_PAUSE_CLOCK (RT) message instructs the active framer to pause the bus. The payload of the
 message contains the restart time (RT), which indicates whether the interruption is to be of a specified time and/or
 phase duration.
- The NEXT_RESET_BUS message instructs all components on the bus to be reset. In this case, all devices on the bus are reset and are disconnected from the bus. Subsequent reconnection to the bus follows the same process as when the bus is first initialized.
- The NEXT_SHUTDOWN_BUS message instructs all devices that the bus is to be shut down.

4.9.5 Data Channel Configuration

This section describes the SLIMbus messages associated with configuring a SLIMbus data channel. Note that the manager device is responsible for allocating the available bandwidth as required for each data channel.

- The CONNECT_SOURCE (PN, EP, CN) and CONNECT_SINK (PN, EP, CN) messages are issued to the
 respective devices, defining the ports between which a data channel is to be established. The end point parameter
 allows up to eight channels to share the same port number. Multiple destinations (sinks) can be configured for a
 channel, if required. The payload of each message contains the port number (PN), end point (EP), and the channel
 number (CN) parameters.
- The BEGIN_RECONFIGURATION message is issued to define a Reconfiguration Boundary point: subsequent NEXT_* messages become active at the first valid superframe boundary following receipt of the RECONFIGURE_ NOW message. A valid boundary must be at least two slots after the end of the RECONFIGURE_NOW message.
- The NEXT_DEFINE_CHANNEL (CN, TP, SD, SL, LN) message informs the associated devices of the structure of the data channel. The payload of this message contains the channel number (CN), TP, SD, SL, and LN parameters for the data channel.
- The NEXT_DEFINE_CONTENT (CN, FL, PR, AF, DT, CL, DL), or CHANGE_CONTENT (CN, FL, PR, AF, DT, CL, DL) message provides more detailed information about the data channel contents. The payload of this message contains the channel number (CN), frequency locked (FL), presence rate (PR), auxiliary bits format (AF), data type (DT), channel link (CL), and data length (DL) parameters.
- The NEXT_ACTIVATE_CHANNEL (CN) message instructs the channel to be activated at the next reconfiguration boundary. The payload of this message contains the channel number (CN) only.
- The RECONFIGURE_NOW message completes the reconfiguration sequence, causing all of the NEXT_ messages since the BEGIN_RECONFIGURATION to become active at the next valid superframe boundary. (A valid boundary must be at least two slots after the end of the RECONFIGURE_NOW message.)
- Active channels can be reconfigured using the CHANGE_CONTENT, NEXT_DEFINE_CONTENT, or NEXT_ DEFINE_CHANNEL messages. Note that these changes can be effected without interrupting the data channel; the NEXT_DEFINE_CHANNEL, for example, may be used to change a segment distribution, in order to reallocate the SLIMbus bandwidth.
- An active channel can be paused using the NEXT_DEACTIVATE_CHANNEL message and reinstated using the NEXT_ACTIVATE_CHANNEL message.
- Data channels can be disconnected using the DISCONNECT_PORT or NEXT_REMOVE_CHANNEL messages. These messages provide equivalent functionality, but use different parameters (PN or CN respectively) to identify the affected signal path.

4.10 SLIMbus Interface Control

The CS42L92 features a MIPI-compliant SLIMbus interface. It supports multichannel audio input/output and control register read/write access.

The SLIMbus interface on CS42L92 comprises a generic device, framer device, interface device, and 16 data ports, providing up to 8 input (RX) channels and up to 8 output (TX) channels. Multiline capability is supported, offering additional bandwidth and system-level flexibility.

The interface supports up to eight audio input channels and up to eight audio output channels. Mixed sample rates can be supported simultaneously. The audio signal paths associated with the SLIMbus interface are described in Section 4.3.



The SLIMbus interface also supports read/write access to the CS42L92 control registers via the value map of the interface device, as described in Section 4.10.5.

The SLIMbus clocking rate and channel allocations are controlled by the manager device. The message channel and data channel bandwidth may be dynamically adjusted according to the application requirements. Note that the manager device functions are not implemented on the CS42L92, and these bandwidth allocation requirements are outside the scope of this data sheet.

The SLIMbus interface provides full support for 32-bit data words (input and output). Audio data samples up to 32 bits can be routed to the AIF1, AIF3, SLIMbus, S/PDIF, and DAC output paths. Note that other signal paths and signal-processing blocks within the digital core are limited to 24-bit data length; data samples are truncated to 24-bit length if they are routed through any function that does not support 32-bit data words.

4.10.1 SLIMbus Device Parameters

The SLIMbus interface on the CS42L92 comprises three devices. The enumeration address of each device within the SLIMbus interface is derived from the parameters noted in Table 4-57.

Description	Manufacturer ID	Product Code	Device ID	Instance Value	Enumeration Address
Generic	0x01FA	0x6371	0x00	0x00	01FA_6371_0000
Framer	0x01FA	0x6371	0x55	0x00	01FA_6371_5500
Interface	0x01FA	0x6371	0x7F	0x00	01FA_6371_7F00

Table 4-57. SLIMbus Device Parameters

4.10.2 SLIMbus Message Support

The SLIMbus interface on the CS42L92 supports bus messages as described in Table 4-58 and Table 4-59.

Category	Message Code MC[6:0]	Description	Generic	Framer	Interface
Device Management	0x01	REPORT_PRESENT (DC, DCV)	S	S	S
Messages	0x02	ASSIGN_LOGICAL_ADDRESS (LA)	D	D	D
	0x04	RESET_DEVICE ()	D	D	D
	0x08	CHANGE_LOGICAL_ADDRESS (LA)	D	D	D
	0x09	CHANGE_ARBITRATION_PRIORITY (AP)	—		
	0x0C	REQUEST_SELF_ANNOUNCEMENT ()	D	D	D
	0x0F	REPORT_ABSENT ()	—	_	
Data Channel	0x10	CONNECT_SOURCE (PN, EP, CN)	D	_	
Management	0x11	CONNECT_SINK (PN, EP, CN)	D	_	
Messages	0x14	DISCONNECT_PORT (PN)	D	_	
	0x18	CHANGE_CONTENT (CN, FL, PR, AF, DT, CL, DL)	D		
Information	0x20	REQUEST_INFORMATION (TID, EC)	D	D	D
Management	0x21	REQUEST_CLEAR_INFORMATION (TID, EC, CM)	D	D	D
Messages	0x24	REPLY_INFORMATION (TID, IS)	S	S	S
	0x28	CLEAR_INFORMATION (EC, CM)	D	D	D
	0x29	REPORT_INFORMATION (EC, IS)	—	—	S

Table 4-58. SLIMbus Message Support



Category	Message Code MC[6:0]	Description	Generic	Framer	Interface
Reconfiguration	0x40	BEGIN_RECONFIGURATION ()	D	D	D
Messages	0x44	NEXT_ACTIVE_FRAMER (LAIF, NCo, NCi)	_	D	—
	0x45	NEXT_SUBFRAME_MODE (SM)	_	D	D
	0x46	NEXT_CLOCK_GEAR (CG)	_	D	—
	0x47	NEXT_ROOT_FREQUENCY (RF)	_	D	—
	0x4A	NEXT_PAUSE_CLOCK (RT)	_	D	_
	0x4B	NEXT_RESET_BUS ()	_	D	—
	0x4C	NEXT_SHUTDOWN_BUS ()	_	D	—
	0x50	NEXT_DEFINE_CHANNEL (CN, TP, SD, SL, LN)	D	_	_
	0x51	NEXT_DEFINE_CONTENT (CN, FL, PR, AF, DT, CL, DL)	D	_	—
	0x54	NEXT_ACTIVATE_CHANNEL (CN)	D	_	—
	0x55	NEXT_DEACTIVATE_CHANNEL (CN)	D	_	—
	0x58	NEXT_REMOVE_CHANNEL (CN)	D	_	—
	0x5F	RECONFIGURE_NOW ()	D	D	D
Value Management	0x60	REQUEST_VALUE (TID, EC)	_	_	D
Messages	0x61	REQUEST_CHANGE_VALUE (TID, EC, VU)	-	_	—
	0x64	REPLY_VALUE (TID, VS)	—	_	S
	0x68	CHANGE_VALUE (EC, VU)	_	_	D

Table 4-58. SLIMbus Message Support (Cont.)

Notes:

• S = Supported as a source device only.

• D = Supported as a destination device only.

The CS42L92 SLIMbus component must be reset prior to scheduling a hardware reset or power-on reset. This can be achieved using the RESET_DEVICE message (issued to the CS42L92 interface device), or else using the NEXT_RESET_BUS message.

Additional notes on specific SLIMbus parameters are described in Table 4-59.

Table 4-59.	SLIMbus Parameter Supp	ort

Parameter Code	Description	Comments
AF	Auxiliary Bits Format	_
CG	Clock Gear	-
CL	Channel Link	_
СМ	Clear Mask	CS42L92 does not fully support this function. The CM bytes of the REQUEST_ CLEAR_INFORMATION or CLEAR_INFORMATION messages must not be sent to CS42L92 Devices. When either of these messages is received, all bits within the specified Information Slice are cleared.
CN	Channel Number	-
DC	Device Class	_
DCV	Device Class Variation	_
DL	Data Length	
DT	Data Type	CS42L92 supports the following DT codes:
		0x0 = Not indicated
		0x1 = LPCM audio
		Note that 2's complement PCM can be supported with $DT = 0x0$.
EC	Element Code	—
EP	End Point	
FL	Frequency Locked	—
IS	Information Slice	—
LA	Logical Address	
LAIF	Logical Address, Incoming Framer	_
LN	Data Line	All LN codes (0–7) are supported. The LN value must be equal to one of the data lines that is mapped to one of the CS42L92 SLIMDAT <i>n</i> pins.
		The mapping of the secondary SLIMDAT <i>n</i> pins onto the SLIMbus data lines is configurable, using the value elements of the interface device. (Note that the primary data pin, SLIMDAT1, is always mapped to SLIMbus data line DATA0.)
NCi	Number of Incoming Framer Clock Cycles	
NCo	Number of Outgoing Framer Clock Cycles	-



Parameter Code	Description	Comments
PN	Port Number	Note that the Port Numbers of the CS42L92 SLIMbus paths are
		register-configurable, as described in Table 4-62.
PR	Presence Rate	Note that the Presence Rate must be the same as the sample rate selected for the associated CS42L92 SLIMbus path.
RF	Root Frequency	CS42L92 supports the following RF codes as Active Framer:
		0x1 = 24.576 MHz
		0x2 = 22.5792 MHz
		All codes are supported when CS42L92 is not the Active Framer.
RT	Restart Time	CS42L92 supports the following RT codes:
		0x0 = Fast Recovery
		0x2 = Unspecified Delay
		When either of these values is specified, the CS42L92 resumes toggling the CLK line within four cycles of the CLK line frequency.
SD	Segment Distribution	Note that any audio channels that are assigned the same SAMPLE_RATE_n or ASYNC_SAMPLE_RATE_n value must also be assigned the same Segment Interval.
SL	Segment Length	Note that any active data channels that are assigned the same Port Number must also be assigned the same Segment Length.
SM	Subframe Mode	-
TID	Transaction ID	-
TP	Transport Protocol	CS42L92 supports the following TP codes, according to the applicable audio channel port:
		Audio TX channel ports: 0x0 (Isochronous Protocol) or 0x1 (Pushed Protocol)
		Audio RX channel ports: 0x0 (Isochronous Protocol) or 0x2 (Pulled Protocol)
		Note that any active data channels that are assigned the same Port Number must also be assigned the same Transport Protocol.
VS	Value Slice	—
VU	Value Update	—

Table 4-59. SLIMbus Parameter Support (Cont.)

4.10.3 SLIMbus Clocking Control

The clock frequency of the SLIMbus interface is not fixed, and may be set according to the application requirements. The clock frequency can be reconfigured dynamically as required.

The CS42L92 SLIMbus interface includes a framer device. When configured as the active framer, the SLIMbus clock (SLIMCLK) is an output from the CS42L92. At other times, SLIMCLK is an input. The framer function can be transferred from one device to another; this is known as framer handover, and is controlled by the manager device.

The supported root frequencies in active framer mode are 24.576 or 22.5792 MHz only. At other times, the supported root frequencies are as defined in the MIPI Alliance specification for SLIMbus.

Under normal operating conditions, the SLIMbus interface operates with a fixed root frequency (RF); dynamic updates to the bus rate are applied using a selectable clock gear (CG) function. The root frequency and the clock gear setting are controlled by the manager device; these parameters are transmitted in every SLIMbus superframe to all devices on the bus.

In Gear 10 (the highest clock gear setting), the SLIMCLK input (or output) frequency is equal to the root frequency. In lower gears, the SLIMCLK frequency is reduced by increasing powers of 2.

The clock gear definition is shown in Table 4-60.

Note: The 24.576MHz root frequency is an example only; other frequencies are also supported.

Clock Gear	Description	SLIMCLK Frequency ¹
10	Divide by 1	24.576 MHz
9	Divide by 2	12.288 MHz
8	Divide by 4	6.144 MHz
7	Divide by 8	3.072 MHz

Table 4-60. SLIMbus Clock Gear Selection	able 4-60.	SLIMbus	Clock Gear	Selection
--	------------	---------	-------------------	-----------



Clock Gear	Description	SLIMCLK Frequency ¹		
6	Divide by 16	1.536 MHz		
5	Divide by 32	768 kHz		
4	Divide by 64	384 kHz		
3	Divide by 128	192 kHz		
2	Divide by 256	96 kHz		
1	Divide by 512	48 kHz		

Table 4-60. SLIMbus Clock Gear Selection (Cont.)

1. Assuming 24.576-MHz root frequency

If the CS42L92 is the active framer, the SLIMCLK output is synchronized to the SYSCLK or ASYNCCLK system clock, as selected by the SLIMCLK_SRC bit. The applicable system clock must be enabled and configured at the SLIMbus root frequency, whenever the CS42L92 is the active framer.

If the CS42L92 is not the active framer, the SLIMCLK input can be used to provide a reference source for the FLLs. The frequency of this reference is controlled using SLIMCLK_REF_GEAR, as described in Table 4-61.

The input clock reference for the FLLs is selected by using FLL*n*_REFCLK_SRC. If SLIMbus is selected as the clock source, the reference signal is generated using an adaptive divider on the SLIMCLK input. The divider automatically adapts to the SLIMbus clock gear (CG). If the clock gear on the bus is lower than the SLIMCLK_REF_GEAR, the selected reference frequency cannot be supported, and the SLIMbus clock reference is disabled.

See Section 4.16 for details of system clocking and the FLLs.

Register Address	Bit	Label	Default	Description		
R1507 (0x05E3)	4	SLIMCLK_SRC	0	SLIMbus Clock source		
SLIMbus_Framer_				Selects the SLIMbus reference clock in Active Framer mode.		
Ref_Gear				0 = SYSCLK		
				1 = ASYNCCLK		
				Note that the applicable clock must be enabled, and configured at the SLIMbus Root Frequency, in Active Framer mode.		
	3:0	SLIMCLK_REF_ GEAR[3:0]	0x0	SLIMbus Clock Reference control. Sets the SLIMbus reference clock relative to the SLIMbus Root Frequency (RF).		
				0x0 = Clock stopped 0x4 = Gear 4. (RF/64) 0x8 = Gear 8. (RF/4)		
				0x1 = Gear 1. (RF/512) 0x5 = Gear 5. (RF/32) 0x9 = Gear 9. (RF/2)		
				0x2 = Gear 2. (RF/256) 0x6 = Gear 6. (RF/16) 0xA = Gear 10. (RF)		
				0x3 = Gear 3. (RF/128) 0x7 = Gear 7. (RF/8) All other codes are reserved		

Table 4-61. SLIMbus Clock Reference Control

4.10.4 SLIMbus Audio Channel Control

4.10.4.1 Port Number Control

The CS42L92 SLIMbus interface supports up to eight audio input (RX) channels and up to eight audio output (TX) channels. The port number and end point number for each channel is configurable using the fields described in Table 4-62.

Register Address	Bit	Label	Default	Description
R1490 (0x05D2)	15:8	SLIMRX2_PORT_ADDR[7:0]	0x01	SLIMbus RX Channel n Port number
SLIMbus_RX_Ports0	7:0	SLIMRX1_PORT_ADDR[7:0]	0x00	Bits [7:5] specify the End Point (valid from
R1491 (0x05D3)	15:8	SLIMRX4_PORT_ADDR[7:0]	0x03	0–7). Bits [4:0] specify the Port Number
SLIMbus_RX_Ports1	7:0	SLIMRX3_PORT_ADDR[7:0]	0x02	(valid from 0–31)
R1492 (0x05D4)	15:8	SLIMRX6_PORT_ADDR[7:0]	0x05	
SLIMbus_RX_Ports2	7:0	SLIMRX5_PORT_ADDR[7:0]	0x04	
R1493 (0x05D5)	15:8	SLIMRX8_PORT_ADDR[7:0]	0x07	
SLIMbus_RX_Ports3	7:0	SLIMRX7_PORT_ADDR[7:0]	0x06	

Table 4-62. SLIMbus Audio Port Number Control



Register Address	Bit	Label	Default	Description
R1494 (0x05D6)	15:8	SLIMTX2_PORT_ADDR[7:0]	0x09	SLIMbus TX Channel n Port number
SLIMbus_TX_Ports0	7:0	SLIMTX1_PORT_ADDR[7:0]	0x08	Valid from 0–31
R1495 (0x05D7)	15:8	SLIMTX4_PORT_ADDR[7:0]	0x0B	Bits [7:5] specify the End Point (valid from
SLIMbus_TX_Ports1	7:0	SLIMTX3_PORT_ADDR[7:0]	0x0A	0–7). Bits [4:0] specify the Port Number
R1496 (0x05D8)	15:8	SLIMTX6_PORT_ADDR[7:0]	0x0D	(valid from 0–31)
SLIMbus_TX_Ports2	7:0	SLIMTX5_PORT_ADDR[7:0]	0x0C	
R1497 (0x05D9)	15:8	SLIMTX8_PORT_ADDR[7:0]	0x0F]
SLIMbus_TX_Ports3	7:0	SLIMTX7_PORT_ADDR[7:0]	0x0E]

Table 4-62. SLIMbus Audio Port Number Control (Cont.)

4.10.4.2 Sample-Rate Control

The SLIMbus audio inputs may be selected as input to the digital mixers or signal-processing functions within the CS42L92 digital core. The SLIMbus audio outputs are derived from the respective output mixers.

The sample rate for each SLIMbus channel is configured using SLIMRX*n*_RATE and SLIMTX*n*_RATE—see Table 4-26.

Note that the SLIMbus interface provides simultaneous support for SYSCLK-referenced and ASYNCCLK-referenced sample rates on different channels. For example, 48-kHz and 44.1-kHz SLIMbus audio paths can be simultaneously supported.

Sample-rate conversion is required when routing the SLIMbus paths to any signal chain that is asynchronous or configured for a different sample rate.

4.10.4.3 Signal Path Enable

The SLIMbus interface supports up to eight audio input (RX) channels and up to eight audio output (TX) channels. Each channel can be enabled or disabled using the fields defined in Table 4-63.

Note: SLIMbus audio channels can be supported only when the corresponding ports have been enabled by the manager device (i.e., in addition to setting the respective enable bits). The status bits in Registers R1527 and R1528 indicate the status of each SLIMbus port.

The system clock, SYSCLK, must be configured and enabled before any audio path is enabled. The ASYNCCLK may also be required, depending on the path configuration. See Section 4.16 for details of the system clocks.

	D ''			Description
Register Address	Bit	Label	Default	Description
R1525 (0x05F5)	7	SLIMRX8_ENA	0	SLIMbus RX Channel n Enable
SLIMbus_RX_	6	SLIMRX7_ENA	0	0 = Disabled
Channel_Enable	5	SLIMRX6_ENA	0	1 = Enabled
	4	SLIMRX5_ENA	0	
	3	SLIMRX4_ENA	0	
	2	SLIMRX3_ENA	0	
	1	SLIMRX2_ENA	0	
	0	SLIMRX1_ENA	0	
R1526 (0x05F6)	7	SLIMTX8_ENA	0	SLIMbus TX Channel n Enable
SLIMbus_TX_	6	SLIMTX7_ENA	0	0 = Disabled
Channel_Enable	5	SLIMTX6_ENA	0	1 = Enabled
	4	SLIMTX5_ENA	0	
	3	SLIMTX4_ENA	0	
	2	SLIMTX3_ENA	0]
	1	SLIMTX2_ENA	0	
	0	SLIMTX1_ENA	0	

Table 4-63.	SLIMbus	Signal	Path	Enable
	O E IIII D G O	orginar		



Register Address	Bit	Label	Default	Description
R1527 (0x05F7)	7	SLIMRX8_PORT_STS	0	SLIMbus RX Channel n Port Status
SLIMbus_RX_	6	SLIMRX7_PORT_STS	0	(Read only)
Port_Status	5	SLIMRX6_PORT_STS	0	0 = Disabled
	4	SLIMRX5_PORT_STS	0	1 = Configured and active
	3	SLIMRX4_PORT_STS	0	
	2	SLIMRX3_PORT_STS	0	
	1	SLIMRX2_PORT_STS	0	
	0	SLIMRX1_PORT_STS	0	
R1528 (0x05F8)	7	SLIMTX8_PORT_STS	0	SLIMbus TX Channel n Port Status
SLIMbus_TX_	6	SLIMTX7_PORT_STS	0	(Read only)
Port_Status	5	SLIMTX6_PORT_STS	0	0 = Disabled
	4	SLIMTX5_PORT_STS	0	1 = Configured and active
	3	SLIMTX4_PORT_STS	0	
	2	SLIMTX3_PORT_STS	0	
	1	SLIMTX2_PORT_STS	0	
	0	SLIMTX1_PORT_STS	0	

Table 4-63. SLIMbus Signal Path Enable (Cont.)

4.10.5 SLIMbus Control Register Access

The SLIMbus interface supports read/write access to the CS42L92 control registers via the value map of the interface device. Full read/write access to all registers is possible, via the user value elements portion of the value map.

Note that if multibyte transfers of more than 8 bytes are scheduled (see Section 4.10.5.3), system clocking constraints must be observed to ensure control interface limits are not exceeded. Full details of the applicable clocking requirements are provided in Section 4.16.7.

4.10.5.1 Control Register Write

Register write operations are implemented using the CHANGE_VALUE message. A maximum of two messages may be required, depending on circumstances: the first CHANGE_VALUE message selects the register page (bits [23:8] of the control register address); the second message contains the data and bits [7:0] of the register address. The first message may be omitted if the register page is unchanged from the previous read or write operation.

The required SLIMbus parameters are described in Table 4-64 and Table 4-65, for the generic case of writing the value 0xVVVV to control register address 0xYYYYZZ. Note that it is also possible to write blocks of up to 16 bytes (to consecutive register addresses), as described in Section 4.10.5.3.

Parameter	Value	Description	
Source Address	0xSS	SS is the 8-bit logical address of the message source. This could be any active device on the bus, but is typically the manager device (0xFF).	
Destination Address	0xLL	LL is the 8-bit logical address of the message destination (i.e., the CS42L92 SLIMbus interface device). The value is assigned by the SLIMbus manager device.	
Access Mode	0b1	Selects byte-based access mode.	
Byte Address	0x800	Identifies the user value element for selecting the control register page address.	
Slice Size	0b001	Selects 2-byte slice size	
Value Update	0xYYYY	YYYY is bits [23:8] of the applicable control register address.	

Table 4-64.	Register Write	Message (1)-	-CHANGE_VALUE
-------------	----------------	--------------	---------------

Table 4-65. Register Write Message (2)—CHANGE_VALUE

Parameter	Value	Description
Source Address	0xSS	SS is the 8-bit logical address of the message source. This could be any active device on the bus, but is typically the manager device (0xFF).
Destination Address	0xLL	LL is the 8-bit logical address of the message destination (i.e., the CS42L92 SLIMbus interface device). The value is assigned by the SLIMbus manager device.
Access Mode	0b1	Selects byte-based access mode.



Parameter	Value	Description	
Byte Address	0xUUU	Specifies the value map address, calculated as 0xA00 + (2 x 0xZZ), where ZZ is bits [7:0] of the applicable control register address.	
Slice Size	0b001	Selects 2-byte slice size	
Value Update	0xVVVV	VVVV is the 16-bit data to be written.	

Table 4-65. Register Write Message (2)—CHANGE_VALUE (Cont.)

Note: The first message may be omitted if its contents are unchanged from the previous CHANGE_VALUE message sent to the CS42L92.

4.10.5.2 Control Register Read

Register read operations are implemented using the CHANGE_VALUE and REQUEST_VALUE messages. A maximum of two messages may be required, depending on circumstances: the CHANGE_VALUE message selects the register page (bits [23:8] of the control register address); the REQUEST_VALUE message contains bits [7:0] of the register address. The first message may be omitted if the register page is unchanged from the previous read or write operation.

The required SLIMbus parameters are described in Table 4-66 and Table 4-67, for the generic case of reading the contents of control register address 0xYYYYZZ. Note that it is also possible to read blocks of up to 8 bytes (to consecutive register addresses), as described in Section 4.10.5.3.

Parameter	Value	Description
Source Address	0xSS	SS is the 8-bit logical address of the message source. This could be any active device on the bus, but is typically the manager device (0xFF).
Destination Address	0xLL	LL is the 8-bit logical address of the message destination (i.e., the CS42L92 SLIMbus interface device). The value is assigned by the SLIMbus manager device.
Access Mode	0b1	Selects byte-based access mode.
Byte Address	0x800	Identifies the user value element for selecting the control register page address.
Slice Size	0b001	Selects 2-byte slice size
Value Update	0xYYYY	YYYY is bits [23:8] of the applicable control register address.

Table 4-66. Register Read Message (1)—CHANGE_VALUE

Table 4-67. Register Read Message (2)—REQUEST_VALUE

Parameter	Value	Description
Source Address		SS is the 8-bit logical address of the message source. This could be any active device on the bus, but is typically the manager device (0xFF).
Destination Address		LL is the 8-bit logical address of the message destination (i.e., the CS42L92 SLIMbus interface device). The value is assigned by the SLIMbus manager device.
Access Mode	0b1	Selects byte-based access mode.
Byte Address	0xUUU	Specifies the value map address, calculated as 0xA00 + (2 x 0xZZ), where ZZ is bits [7:0] of the applicable control register address.
Slice Size	0b001	Selects 2-byte slice size
Transaction ID	0xTTTT	TTTT is the 16-bit transaction ID for the message. The value is assigned by the SLIMbus manager device.

Note: The first message may be omitted if its contents are unchanged from the previous CHANGE_VALUE message sent to the CS42L92.

The CS42L92 responds to the register read commands in accordance with the normal SLIMbus protocols.

Note that the CS42L92 assumes that sufficient control space slots are available in which to provide its response before the next REQUEST_VALUE message is received. The CS42L92 response is made using a REPLY_VALUE message; the SLIMbus manager should wait until the REPLY_VALUE message has been received before sending the next REQUEST_VALUE message. If additional REQUEST_VALUE messages are received before the CS42L92 response has been made, the earlier REQUEST_VALUE messages are ignored (i.e., only the last REQUEST_VALUE message is serviced).

4.10.5.3 Multibyte Control Register Access

Register data transfers of up to 16 bytes can be configured using the slice size parameter in the second message of the applicable protocol (see Table 4-65 or Table 4-67). Additional value update words are appended to respective data message in this case, with the applicable data contents.



Multibyte register read/write access is supported with slice size of 2, 4, 8, 12, or 16 bytes. Note that if multibyte transfers of more than 8 bytes are scheduled, system clocking constraints must be observed to ensure control interface limits are not exceeded. See Section 4.16.7.

When a 2-byte transfer is selected, the register address 0xYYYYZZ is used (using the same naming conventions as above). When more than 2 bytes are transferred, the register address is automatically incremented as described in Table 4-68.

Register Address (<0x3000)	Byte Sequence
Base address (0xYYYYZZ)	Bytes 2 and 1 (0xVVVV)
Base address + 1	Bytes 4 and 3
Base address + 2	Bytes 6 and 5
Base address + 3	Bytes 8 and 7
Base address + 4	Bytes 10 and 9
Base address + 5	Bytes 12 and 11
Base address + 6	Bytes 14 and 13
Base address + 7	Bytes 16 and 15

Table 4-68. SLIMbus Register Read/Write Sequence—16-Bit Register Space (< 0x3000)

Note: Register addresses from R12288 (0x3000) upwards are formatted as 32-bit words. When accessing these addresses, the slice size should be a multiple of 4 bytes and the byte address should be aligned with the 32-bit data word boundaries (i.e., an even number). The byte ordering for these register addresses is described in Table 4-69.

Table 4-69. SLIMbus Register Read/Write Sequence—32-Bit Register Space (≥ 0x3000)

Register Address (≥0x3000)	Byte Sequence
Base address (0xYYYYZZ)	Bytes 4, 3, 2, 1
Base address + 2	Bytes 8, 7, 6, 5
Base address + 4	Bytes 12, 11, 10, 9
Base address + 6	Bytes 16, 15, 14, 13

4.11 Output Signal Path

The CS42L92 provides four stereo pairs of audio output signal paths. These outputs comprise ground-referenced headphone drivers, and a digital output interface suitable for external speaker drivers. The output signal paths are summarized in Table 4-70.

Signal Path	Descriptions	Output Pins
OUT1L, OUT1R	Ground-referenced headphone/earpiece output	HPOUT1L, HPOUT1R
OUT2L, OUT2R	Ground-referenced headphone/earpiece output	HPOUT2L, HPOUT2R
OUT3L, OUT3R	Ground-referenced headphone/earpiece output	HPOUT3L, HPOUT3R or HPOUT4L, HPOUT4R
OUT5L, OUT5R	Digital speaker (PDM) output	SPKDAT, SPKCLK

Table 4-70.	Output Signal Pa	ath Summary
-------------	------------------	-------------

The analog output paths incorporate high performance 32-bit sigma-delta DACs.

Under default conditions, the headphone drivers provide a stereo, single-ended output. A mono mode is also supported, providing a differential (BTL) output configuration. The ground-referenced headphone output paths incorporate a common mode feedback path for rejection of system-related noise. These outputs support direct connection to headphone loads, with no requirement for AC coupling capacitors.

The OUT1 and OUT2 paths can be configured in High Performance Mode, offering exceptional performance on the respective headphone outputs. Balanced stereo headphone loads can be supported by assigning the OUT1 and OUT2 signal paths for the left and right channels respectively, with the associated drivers configured for differential output.



The digital output path provides a stereo pulse-density modulation (PDM) output interface, for connection to external audio devices. A total of two digital output channels are provided.

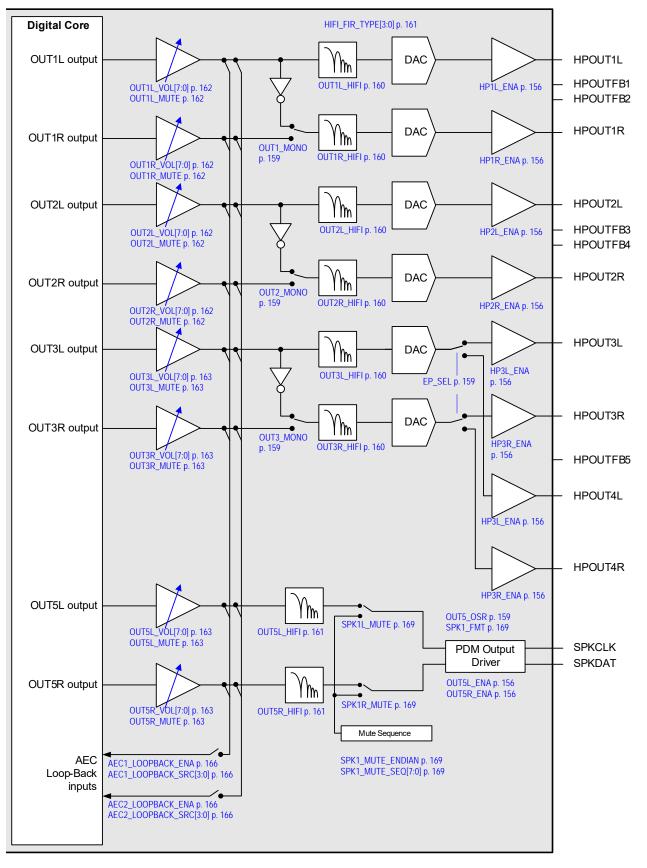
Digital filters can be enabled in the output signal paths, supporting audiophile-quality DAC playback options. These hi-fi filters allow user selection of the preferred characteristics, (e.g., linear phase, antialiasing, or apodizing filter responses).

Digital volume control is available on all outputs (analog and digital), with programmable ramp control for smooth, glitch-free operation. A configurable noise-gate function is available on each output signal path. Any two of the output signal paths may be selected as input to the AEC loop-back paths.

The CS42L92 provides short-circuit detection on the headphone output paths. See Section 4.21 for further details.

The CS42L92 output signal paths are shown in Fig. 4-52. Note that the OUT4 path is not implemented on this device.









4.11.1 Output Signal Path Enable

The output signal paths are enabled using the bits described in Table 4-71. The respective bits must be enabled for analog or digital output on the respective output paths.

The OUT3 path is associated with the HPOUT3 and HPOUT4 output drivers. The HP3L_ENA and HP3R_ENA bits control either the HPOUT3 or HPOUT4 drivers, depending on the EP_SEL register bit selection. See Table 4-74 for details of the EP_SEL register bit.

The output signal paths are muted by default. It is recommended that deselecting the mute should be the final step of the path enable control sequence. Similarly, the mute should be selected as the first step of the path disable control sequence. The output signal path mute functions are controlled using the bits described in Table 4-71.

The supply rails for the analog outputs (HPOUT1–HPOUT4) are generated using an integrated dual-mode charge pump, CP1. The charge pump is enabled automatically by the CS42L92 when required by the output drivers; see Section 4.19.

The CS42L92 schedules a pop-suppressed control sequence to enable or disable the HPOUT1–HPOUT4 signal paths. This is automatically managed by the control-write sequencer in response to setting the respective HP*nx*_ENA bits; see Section 4.18 for further details.

The output signal path enable/disable control sequences are inputs to the interrupt circuit and can be used to trigger an interrupt event when a sequence completes; see Section 4.15.

The system clock, SYSCLK, must be configured and enabled before any audio path is enabled. The ASYNCCLK may also be required, depending on the path configuration. See Section 4.16 for details of the system clocks.

The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the output signal paths and associated DACs. If the frequency is too low, an attempt to enable an output signal path fails. Note that active signal paths are not affected under such circumstances.

The status bits in Register R1025 and R1030 indicate the status of each output signal path. If an underclocked error condition occurs, these bits indicate which signal paths have been enabled.



Register Address	Bit	Label	Default	Description				
R1024 (0x0400)	9	OUT5L_ENA	0	Output Path 5 (left) enable				
Output_Enables_1				0 = Disabled				
				1 = Enabled				
	8	OUT5R_ENA	0	Output Path 5 (right) enable				
				0 = Disabled				
				1 = Enabled				
	5	HP3L_ENA	0	Output Path 3 (left) enable				
		_		If EP SEL = 0, this bit controls the HPOUT3L output driver.				
				If $EP_SEL = 1$, this bit controls the HPOUT4L output driver.				
				0 = Disabled				
				1 = Enabled				
	4	HP3R ENA	0	Output Path 3 (right) enable				
		-		If EP SEL = 0, this bit controls the HPOUT3R output driver.				
				If EP SEL = 1, this bit controls the HPOUT4R output driver.				
				0 = Disabled				
				1 = Enabled				
	3	HP2L ENA	0	Output Path 2 (left) enable				
	-		-					
	2	HP2R ENA	0					
	-		Ŭ					
	1	HP1L ENA	0					
	•		Ũ					
	0	HP1R ENA	0					
	Ŭ		0 = Disabled 1 = Enabled 22R_ENA 0 0 = Disabled 1 = Enabled 1 = Enabled 1 = Enabled 1 = Enabled					
R1025 (0x0401)	9	OUT51 ENA STS	0					
Output_Status_1	Ŭ		Ũ					
output_outdo_1								
	8	OUT5R ENA STS	0					
	Ŭ		Ŭ					
R1030 (0x0406)	5	OUT31 ENA STS	0					
Raw Output Status 1	Ŭ		Ŭ	0 = Disabled				
				1 = Enabled				
	Δ	OUT3R_ENA_STS	0	Output Path 3 (right) enable status				
	-		Ŭ	0 = Disabled				
				1 = Enabled				
	3	OUT2L ENA STS	0	Output Path 2 (left) enable status				
	0		U	0 = Disabled				
				1 = Enabled				
	2	OUT2R ENA STS	0	Output Path 2 (right) enable status				
	2		U	0 = Disabled				
				1 = Enabled				
	1	OUT1L_ENA_STS	0	Output Path 1 (left) enable status				
	'		0	0 = Disabled				
				1 = Enabled				
	0	OUT1R_ENA_STS	0	Output Path 1 (right) enable status				
		OUTIN_ENA_313	U					
				0 = Disabled				
	<u> </u>			1 = Enabled				

Table 4-71. Output Signal Path Enable



4.11.2 Output Signal Path Sample-Rate Control

The output signal paths are derived from the respective output mixers within the CS42L92 digital core. The sample rate for the output signal paths is configured using OUT_RATE—see Table 4-26.

Note that sample-rate conversion is required when routing the output signal paths to any signal chain that is asynchronous or configured for a different sample rate.

A clocking signal is required by the DACs and output signal path circuits. This clock can be provided internally from the SYSCLK or ASYNCCLK system clocks, or can be provided externally via the MCLK*n* pins. The clock source is selected using OUT_CLK_SRC.

The output signal path clock must be synchronized with whichever clock domain is associated with the OUT_RATE setting.

- If OUT_RATE is configured for one of the SYSCLK-related sample rates (SAMPLE_RATE_n), then ASYNCCLK is
 not a valid selection for OUT_CLK_SRC.
- If OUT_RATE is configured for one of the ASYNCCLK-related sample rates (ASYNC_SAMPLE_RATE_n), then SYSCLK is not a valid selection for OUT_CLK_SRC.
- If MCLK*n* is selected as the clock source (OUT_CLK_SRC = 100–110), the respective MCLK frequency must be an integer multiple of the sample rate selected by OUT_RATE. The required clock frequency is either 6.144 MHz (for 48-kHz-related sample rates) or 5.6448 MHz (44.1-kHz-related sample rates). A configurable divider is provided in order to provide flexibility in matching the MCLK input with the required frequency.

If a suitable external MCLK source is available, then the choice exists between selecting the internal SYSCLK/ASYNCCLK or the external MCLK as the output path clock. If the external MCLK source is high quality, then MCLK is the recommended source. If the MCLK source is not high quality, then the SYSCLK/ASYNCCLK source (generated by one of the FLLs) may provide a better clock.

The clocking configuration for the output signal path is shown in Fig. 4-53.

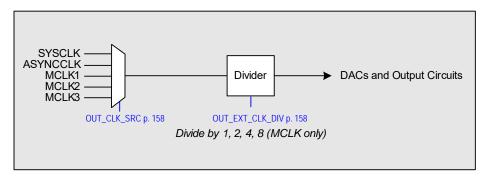


Figure 4-53. Output Signal Path Clocking Control

The output signal path clocking control registers are defined in Table 4-72. The OUT_RATE field is defined in Table 4-26.



Register Address	Bit	Label	Default	Description
R1032 (0x0408)	5:4	OUT_EXT_CLK_DIV	00	Output Signal Path Clock Divider
Output_Rate_1				(only valid if OUT_CLK_SRC selects MCLK <i>n</i> as the clock source)
				00 = Divide by 1
				01 = Divide by 2
				10 = Divide by 4
				11 = Divide by 8
	2:0	OUT_CLK_SRC	000	Output Signal Path Clock Source
				000 = SYSCLK
				001 = ASYNCCLK
				100 = MCLK1
				101 = MCLK2
				110 = MCLK3
				All other codes are reserved
				Note that the selected clock must be synchronized with whichever clock domain is associated with the OUT_RATE setting.

Table 4-72. Output Signal Path Clocking Control

4.11.3 Output Signal Path Control

The OUT1 and OUT2 paths support selectable configuration for low-power or high-performance operation. The applicable mode is selected using the CP_DAC_MODE bit, as described in Table 4-74. The High Performance Mode is configured by default.

Under default register conditions, the headphone output (HPOUT1–HPOUT4) paths are configured for stereo output. The headphone paths can be configured for mono differential (BTL) output using the OUT*n*_MONO bits; this is ideal for driving an earpiece or hearing aid coil.

If the OUT*n*_MONO bit is set, the respective right channel output is an inverted copy of the left channel output signal; this creates a differential output between the respective outputs. The left and right channel output drivers must both be enabled in Mono Mode; both channels should be enabled simultaneously using the bits described in Table 4-71.

The mono (BTL) signal paths are shown in Fig. 4-52. Note that, in Mono Mode, the effective gain of the signal path is increased by 6 dB.

Note that the EP_SEL and OUT3_MONO bits should not be changed if the HPOUT3 or HPOUT4 drivers are enabled. These bits should be configured before enabling the respective drivers, and should remain unchanged until after the drivers have been disabled. The HPOUT3 and HPOUT4 drivers are enabled using the HP3L_ENA and HP3R_ENA bits, as described in Table 4-71.

The SPKCLK frequency of the PDM output path (OUT5) is controlled by OUT5_OSR, as described in Table 4-73. When the OUT5_OSR bit is set, the audio performance is improved, but power consumption is also increased.

Note that the SPKCLK frequencies noted in Table 4-73 assume that the SYSCLK frequency is a multiple of 6.144 MHz (SYSCLK_FRAC=0). If the SYSCLK frequency is a multiple of 5.6448 MHz (SYSCLK_FRAC=1), the SPKCLK frequency is scaled accordingly.

Condition	SPKCLK Frequency
OUT5_OSR = 0	3.072 MHz
OUT5_OSR = 1	6.144 MHz

The output signal path control registers are defined in Table 4-74.



Register Address	Bit	Label	Default	Description
R1024 (0x0400)	15	EP_SEL	0	Output Path 3 Output Driver select
Output_Enables_1				0 = HPOUT3L and HPOUT3R
				1 = HPOUT4L and HPOUT4R
R1032 (0x408)	6	CP_DAC_MODE	1	Output Path 1 and 2 Performance Mode select
Output_Rate_1				0 = Low Power Mode
				1 = High Performance Mode
R1040 (0x0410)	12	OUT1_MONO	0	Output Path 1 Mono Mode
Output_Path_				(Configures HPOUT1L and HPOUT1R as a mono differential output.)
Config_1L				0 = Disabled
				1 = Enabled
				The gain of the signal path is increased by 6 dB in differential (mono) mode.
R1048 (0x0418)	12	OUT2_MONO	0	Output Path 2 Mono Mode
Output_Path_				(Configures HPOUT2L and HPOUT2R as a mono differential output.)
Config_2L				0 = Disabled
				1 = Enabled
				The gain of the signal path is increased by 6 dB in differential (mono) mode.
R1056 (0x0420)	12	OUT3_MONO	0	Output Path 3 Mono Mode
Output_Path_				(Configures HPOUT3 and HPOUT4 as a mono differential output.)
Config_3L				0 = Disabled
				1 = Enabled
				The gain of the signal path is increased by 6 dB in differential (mono) mode.
R1072 (0x0430)	13	OUT5_OSR	0	Output Path 5 Oversample Rate
Output_Path_				0 = Normal mode
Config_5L				1 = High Performance mode

Table 4-74. Output Signal Path Control

4.11.4 Output Signal Path Digital Filter Control

An integrated signal-processing engine on the CS42L92 supports digital filter requirements for range of hi-fi applications. Preset filter coefficients are held in on-board ROM, and can be configured and enabled as required. The hi-fi filters are tailored to specific sample rates, and provide options relating to passband frequency, stopband attenuation, and phase-response characteristics.

The filter type is selected using the HIFI_FIR_TYPE field. The available filters are each described in Table 4-75. The digital filter can be enabled in any output path, using the respective OUT*nx*_HIFI bits.

Note that only one filter type can be selected at any time, but the applicable filter can be enabled on any number of output paths simultaneously. The supported sample rates for each filter type are noted in Table 4-75—the selected filter must be consistent with the output sample rate (OUT_RATE) setting.

The digital filter can be enabled or disabled independently in any output path. A short interruption to the playback if the filter type is changed while the hi-fi filters are enabled on any output path.

The hi-fi digital filters are described in Table 4-75.

Туре	Sample Rate	Description	Passband	Stopband	Passband Ripple	Stopband Attenuation
0	48 kHz, 96 kHz,	Deep stopband, linear phase	0.454 fs	0.546 fs	0.001 dB	120 dB
1	192 kHz, 44.1 kHz, 88.2 kHz, 176.4 kHz	Deep stopband, minimum phase	0.454 fs	0.546 fs	0.001 dB	120 dB
2	48 kHz, 44.1 kHz	Antialias, linear phase	0.417 fs	0.5 fs	0.001 dB	110 dB
3		Antialias, minimum phase	0.417 fs	0.5 fs	0.001 dB	110 dB
4	96 kHz, 192 kHz,	Non-apodizing, linear phase	0.227 fs	0.5 fs	0.001 dB	120 dB
5	384 kHz, 88.2 kHz,	Non-apodizing, minimum phase	0.227 fs	0.5 fs	0.001 dB	120 dB
6	176.4 kHz, 352.8 kHz	Apodizing, linear phase	0.227 fs	0.454 fs	0.001 dB	120 dB
7	552.0 KI IZ	Apodizing, minimum phase	0.227 fs	0.454 fs	0.001 dB	120 dB

Table 4-75. Output Signal Path Digital Filter Types



Туре	Sample Rate	Description	Passband	Stopband	Passband Ripple	Stopband Attenuation
8	192 kHz, 384 kHz,	Non-apodizing, linear phase	0.114 fs	0.5 fs	0.001 dB	125 dB
9	176.4 kHz,	Non-apodizing, minimum phase	0.114 fs	0.5 fs	0.001 dB	125 dB
10	352.8 kHz	Apodizing, linear phase	0.114 fs	0.454 fs	0.001 dB	125 dB
11		Apodizing, minimum phase	0.114 fs	0.454 fs	0.001 dB	125 dB
12	192 kHz, 384 kHz,	Non-apodizing, non-interpolating linear phase	0.114 fs	0.499 fs	0.001 dB	125 dB
13	176.4 kHz,	Non-apodizing, non-interpolating minimum phase	0.114 fs	0.499 fs	0.001 dB	125 dB
14	352.8 kHz	Apodizing, non-interpolating linear phase	0.114 fs	0.499 fs	0.001 dB	125 dB
15		Apodizing, non-interpolating minimum phase	0.114 fs	0.499 fs	0.001 dB	125 dB

Table 4-75. Output Signal Path Digital Filter Types (Cont.)

At 48 kHz (or 44.1 kHz) sample rate, the key parameters of the hi-fi digital filters are the stopband attenuation and phase response. The stopband characteristics are noted in Table 4-75.

The choice between linear phase and minimum phase filters determines time-domain effects of the filter transfer function. Linear phase offers zero group delay, and equal levels of pre- and postringing. Minimum phase offers minimum preringing and latency, but higher levels of postringing and group delay distortion.

For sample rates above 48kHz, a choice between apodizing and non-apodizing filters is available. Apodizing filters have the capability to reduce time-domain distortion—this can be used to eliminate time smear effects introduced by other filters in the signal chain, provided the other filters have a flat frequency response throughout the apodizing filter's cut-off region. The cut-off (stopband) frequency of the apodizing filters are slightly lower than the respective non-apodizing filter response.

The hi-fi digital filter control registers are described in Table 4-76.

Register Address	Bit	Label	Default	Description
R1040 (0x0410)	14	OUT1L_HIFI	0	Output Path 1 (Left) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_1L				1 = Enabled
R1044 (0x0414)	14	OUT1R_HIFI	0	Output Path 1 (Right) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_1R				1 = Enabled
R1048 (0x0418)	14	OUT2L_HIFI	0	Output Path 2 (Left) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_2L				1 = Enabled
R1052 (0x041C)	14	OUT2R_HIFI	0	Output Path 2 (Right) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_2R				1 = Enabled
R1056 (0x0420)	14	OUT3L_HIFI	0	Output Path 3 (Left) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_3L				1 = Enabled
R1060 (0x0424)	14	OUT3R_HIFI	0	Output Path 3 (Right) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_3R				1 = Enabled

 Table 4-76. Output Signal Path Digital Filter Control



Register Address	Bit	Label	Default	Description
R1072 (0x0430)	14	OUT5L_HIFI	0	Output Path 5 (Left) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_5L				1 = Enabled
R1076 (0x0434)	14	OUT5R_HIFI	0	Output Path 5 (Right) Hi-Fi Filter Enable
Output_Path_				0 = Disabled
Config_5R				1 = Enabled
R1102 (0x044E)	3:0	HIFI_FIR_	0x0	Output Path Hi-Fi Filter Select
Filter_Control		TYPE[3:0]		0x0 = 192 kHz deep stopband, linear phase
				0x1 = 192 kHz deep stopband, minimum phase
				0x2 = 48 kHz antialias, linear phase
				0x3 = 48 kHz antialias, minimum phase
				0x4 = 384 kHz non-apodizing, linear phase
				0x5 = 384 kHz non-apodizing, minimum phase
				0x6 = 384 kHz apodizing, linear phase
				0x7 = 384 kHz apodizing, minimum phase
				0x8 = 384 kHz non-apodizing, linear phase
				0x9 = 384 kHz non-apodizing, minimum phase
				0xA = 384 kHz apodizing, linear phase
				0xB = 384 kHz apodizing, minimum phase
				0xC = 384 kHz non-apodizing, non-interpolating linear phase
				0xD = 384 kHz non-apodizing, non-interpolating minimum phase
				0xE = 384 kHz apodizing, non-interpolating linear phase
				0xF = 384 kHz apodizing, non-interpolating minimum phase

Table 4-76. Output Signal Path Digital Filter Control (Cont.)

4.11.5 Output Signal Path Digital Volume Control

A digital volume control is provided on each output signal path, providing –64 to +31.5 dB gain control in 0.5-dB steps. An independent mute control is also provided for each output signal path.

Whenever the gain or mute setting is changed, the signal path gain is ramped up or down to the new settings at a programmable rate. For increasing gain (or unmute), the rate is controlled by OUT_VI_RAMP. For decreasing gain (or mute), the rate is controlled by OUT_VD_RAMP.

Note: The OUT_VI_RAMP and OUT_VD_RAMP fields should not be changed while a volume ramp is in progress.

The OUT_VU bits control the loading of the output signal path digital volume and mute controls. When OUT_VU is cleared, the digital volume and mute settings are loaded into the respective control register, but do not change the signal path gain. The digital volume and mute settings on all of the output signal paths are updated when a 1 is written to OUT_VU. This makes it possible to update the gain of multiple signal paths simultaneously.

Note that, although the digital-volume controls provide 0.5-dB steps, the internal circuits provide signal gain adjustment in 0.125-dB steps. This allows a very high degree of gain control—smooth volume ramping under all operating conditions.

Note: The 0 dBFS level of the OUT5 digital output path is not equal to the 0 dBFS level of the CS42L92 digital core. The maximum digital output level is –6 dBFS (see Table 3-8). Under 0 dB gain conditions, a 0 dBFS output from the digital core corresponds to a –6 dBFS level in the PDM output.

The digital volume control registers are described in Table 4-77 and Table 4-78.



Register Address	Bit	Label	Default		Description		
R1033 (0x0409)	6:4	OUT_VD_	010	Output Volume Decreasir		(6 dB)	
Output Volume		RAMP[2:0]		This field should not be c			
Ramp				000 = 0 ms	011 = 2 ms	110 = 15 ms	
				001 = 0.5 ms	100 = 4 ms	111 = 30 ms	
				010 = 1 ms	101 = 8 ms		
	2:0	OUT_VI_	010	Output Volume Increasing	g Ramp Rate (seconds/6	dB)	
		RAMP[2:0]		This field should not be c	hanged while a volume r	amp is in progress.	
				000 = 0 ms	011 = 2 ms	110 = 15 ms	
				001 = 0.5 ms	100 = 4 ms	111 = 30 ms	
				010 = 1 ms	101 = 8 ms		
R1041 (0x0411) DAC Digital	9	OUT_VU	See Footnote 1	Output Signal Paths Volu Paths Volume and Mute		his bit causes the Output Signal multaneously	
Volume_1L	8	OUT1L_MUTE	1	Output Path 1 (Left) Digit	tal Mute		
				0 = Unmute			
				1 = Mute			
	7:0	OUT1L_VOL[7:0]	0x80		•	8 for volume register definition).	
				-64 dB to +31.5 dB in 0.5	5-dB steps		
				0x00 = -64dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved	
				0x01 = -63.5dB	… (0.5-dB steps)		
				(0.5-dB steps)	0xBF = +31.5 dB		
R1045 (0x0415)	9	OUT_VU	See Footnote 1	Paths Volume and Mute s	me Update. Writing 1 to the	his bit causes the Output Signal	
DAC_Digital_ Volume_1R	8	OUT1R MUTE	1	Output Path 1 (Right) Dig		Inditalleously	
volume_nt	0		•	0 = Unmute	gital Mate		
				1 = Mute			
_	7:0	OUT1R_VOL[7:0]	0x80	Output Path 1 (Right) Digital Volume (see Table 4-78 for volume register definition).			
				-64 dB to +31.5 dB in 0.5-dB steps			
				0x00 = -64 dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved	
				0x01 = -63.5 dB	(0.5-dB steps)		
				(0.5-dB steps)	0xBF = +31.5 dB		
R1049 (0x0419)	9	OUT VU	See			his bit causes the Output Signal	
DAC Digital		_	Footnote 1	Paths Volume and Mute			
Volume_2L	8	OUT2L_MUTE	1	Output Path 2 (Left) Digit	tal Mute		
				0 = Unmute			
				1 = Mute			
	7:0	OUT2L_VOL[7:0]	0x80		•	8 for volume register definition).	
				-64 dB to +31.5 dB in 0.5			
				0x00 = -64 dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved	
				0x01 = -63.5 dB	(0.5-dB steps)		
				(0.5-dB steps)	0xBF = +31.5 dB		
R1053 (0x041D)	9	OUT_VU	See	Paths Volume and Mute s	me Update. Writing 1 to the	his bit causes the Output Signal	
DAC_Digital_ Volume_2R	8	OUT2R MUTE	Footnote 1	Output Path 2 (Right) Dig		munaneousiy	
VOIUITIE_2R	0			0 = Unmute			
				1 = Mute			
	7:0	OUT2R_VOL[7:0]	0x80	Output Path 2 (Right) Dig	gital Volume (see Table 4	-78 for volume register	
				definition). 64 dB to ± 21.5 dB in 0.6	5 dB stops		
				-64 dB to +31.5 dB in 0.5 0x00 = -64 dB	0x80 = 0 dB	0xC0 to 0xFF = Reserved	
				0x00 = -64 dB 0x01 = -63.5 dB		UXCU IU UXFF = Reserved	
					(0.5-dB steps)		
				… (0.5-dB steps)	0xBF = +31.5 dB		

Table 4-77. Output Signal Path Digital Volume Control



Register Address	Bit	Label	Default	Description
R1057 (0x0421)	9	OUT_VU	See	Output Signal Paths Volume Update. Writing 1 to this bit causes the Output Signal
DAC_Digital_			Footnote 1	Paths Volume and Mute settings to be updated simultaneously
Volume_3L	8	OUT3L_MUTE	1	Output Path 3 (Left) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	OUT3L_VOL[7:0]	0x80	Output Path 3 (Left) Digital Volume (see Table 4-78 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB 0x80 = 0 dB 0xC0 to 0xFF = Reserved
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB
R1061 (0x0425)	9	OUT_VU	See	Output Signal Paths Volume Update. Writing 1 to this bit causes the Output Signal
DAC_Digital_			Footnote 1	Paths Volume and Mute settings to be updated simultaneously
Volume_3R	8	OUT3R_MUTE	1	Output Path 3 (Right) Digital Mute
				0 = Unmute
			0.00	1 = Mute
	7:0	OUT3R_VOL[7:0]	0x80	Output Path 3 (Right) Digital Volume (see Table 4-78 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB 0x80 = 0 dB 0xC0 to 0xFF = Reserved
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB
R1073 (0x0431) DAC_Digital_	9	OUT_VU	See Footnote 1	Output Signal Paths Volume Update. Writing 1 to this bit causes the Output Signal Paths Volume and Mute settings to be updated simultaneously
Volume_5L	8	OUT5L_MUTE	1	Output Path 5 (Left) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	OUT5L_VOL[7:0]	0x80	Output Path 5 (Left) Digital Volume (see Table 4-78 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB 0x80 = 0 dB 0xC0 to 0xFF = Reserved
				0x01 = -63.5 dB (0.5-dB steps)
				(0.5-dB steps) 0xBF = +31.5 dB
R1077 (0x0435) DAC Digital	9	OUT_VU	See Footnote 1	Output Signal Paths Volume Update. Writing 1 to this bit causes the Output Signal Paths Volume and Mute settings to be updated simultaneously
Volume_5R	8	OUT5R_MUTE	1	Output Path 5 (Right) Digital Mute
				0 = Unmute
				1 = Mute
	7:0	OUT5R_VOL[7:0]	0x80	Output Path 5 (Right) Digital Volume (see Table 4-78 for volume register definition).
				–64 dB to +31.5 dB in 0.5-dB steps
				0x00 = -64 dB $0x80 = 0 dB$ $0xC0 to 0xFF = Reserved$
				0x01 = -63.5 dB (0.5-dB steps)
				$(0.5-dB \text{ steps})$ 0xBF = +31.5 dB
		1		

Table 4-77. Output Signal Path Digital Volume Control (Cont.)

1. Default is not applicable to these write-only bits

Table 4-78 lists the output signal path digital volume settings.

Output Volume Register	Volume (dB)						
0x00	-64.0	0x31	-39.5	0x62	-15.0	0x93	9.5
0x01	-63.5	0x32	-39.0	0x63	-14.5	0x94	10.0
0x02	-63.0	0x33	-38.5	0x64	-14.0	0x95	10.5
0x03	-62.5	0x34	-38.0	0x65	-13.5	0x96	11.0
0x04	-62.0	0x35	-37.5	0x66	-13.0	0x97	11.5
0x05	-61.5	0x36	-37.0	0x67	-12.5	0x98	12.0
0x06	-61.0	0x37	-36.5	0x68	-12.0	0x99	12.5
0x07	-60.5	0x38	-36.0	0x69	-11.5	0x9A	13.0



Output Volume Register	Volume (dB)						
0x08	-60.0	0x39	-35.5	0x6A	-11.0	0x9B	13.5
0x09	-59.5	0x3A	-35.0	0x6B	-10.5	0x9C	14.0
0x0A	-59.0	0x3B	-34.5	0x6C	-10.0	0x9D	14.5
0x0B	-58.5	0x3C	-34.0	0x6D	-9.5	0x9E	15.0
0x0C	-58.0	0x3D	-33.5	0x6E	-9.0	0x9F	15.5
0x0D	-57.5	0x3E	-33.0	0x6F	-8.5	0xA0	16.0
0x0E	-57.0	0x3F	-32.5	0x70	-8.0	0xA1	16.5
0x0F	-56.5	0x40	-32.0	0x71	-7.5	0xA2	17.0
0x10	-56.0	0x41	-31.5	0x72	-7.0	0xA3	17.5
0x11	-55.5	0x42	-31.0	0x73	-6.5	0xA4	18.0
0x12	-55.0	0x43	-30.5	0x74	-6.0	0xA5	18.5
0x13	-54.5	0x44	-30.0	0x75	-5.5	0xA6	19.0
0x14	-54.0	0x45	-29.5	0x76	-5.0	0xA7	19.5
0x15	-53.5	0x46	-29.0	0x77	-4.5	0xA8	20.0
0x16	-53.0	0x47	-28.5	0x78	-4.0	0xA9	20.5
0x17	-52.5	0x48	-28.0	0x79	-3.5	0xAA	21.0
0x18	-52.0	0x49	-27.5	0x7A	-3.0	0xAB	21.5
0x19	-51.5	0x4A	-27.0	0x7B	-2.5	0xAC	22.0
0x1A	-51.0	0x4B	-26.5	0x7C	-2.0	0xAD	22.5
0x1B	-50.5	0x4C	-26.0	0x7D	-1.5	0xAE	23.0
0x1C	-50.0	0x4D	-25.5	0x7E	-1.0	0xAF	23.5
0x1D	-49.5	0x4E	-25.0	0x7F	-0.5	0xB0	24.0
0x1E	-49.0	0x4F	-24.5	0x80	0.0	0xB1	24.5
0x1F	-48.5	0x50	-24.0	0x81	0.5	0xB2	25.0
0x20	-48.0	0x51	-23.5	0x82	1.0	0xB3	25.5
0x21	-47.5	0x52	-23.0	0x83	1.5	0xB4	26.0
0x22	-47.0	0x53	-22.5	0x84	2.0	0xB5	26.5
0x23	-46.5	0x54	-22.0	0x85	2.5	0xB6	27.0
0x24	-46.0	0x55	-21.5	0x86	3.0	0xB7	27.5
0x25	-45.5	0x56	-21.0	0x87	3.5	0xB8	28.0
0x26	-45.0	0x57	-20.5	0x88	4.0	0xB9	28.5
0x27	-44.5	0x58	-20.0	0x89	4.5	0xBA	29.0
0x28	-44.0	0x59	-19.5	0x8A	5.0	0xBB	29.5
0x29	-43.5	0x5A	-19.0	0x8B	5.5	0xBC	30.0
0x2A	-43.0	0x5B	-18.5	0x8C	6.0	0xBD	30.5
0x2B	-42.5	0x5C	-18.0	0x8D	6.5	0xBE	31.0
0x2C	-42.0	0x5D	-17.5	0x8E	7.0	0xBF	31.5
0x2D	-41.5	0x5E	-17.0	0x8F	7.5	0xC0–0xFF	Reserved
0x2E	-41.0	0x5F	-16.5	0x90	8.0		1
0x2F	-40.5	0x60	-16.0	0x91	8.5		
0x30	-40.0	0x61	-15.5	0x92	9.0		

Table 4-78. Output Signal Path Digital Volume Range (Cont.)

4.11.6 Output Signal Path Noise-Gate Control

The CS42L92 provides a digital noise-gate function for each output signal path. The noise gate ensures best noise performance when the signal path is idle. When the noise gate is enabled, and the applicable signal level is below the noise-gate threshold, the noise gate is activated, causing the signal path to be muted.

The noise-gate function is enabled by setting NGATE_ENA, as described in Table 4-79.

For each output path, the noise gate may be associated with one or more of the signal path threshold detection functions using the x_NGATE_SRC fields. When more than one signal threshold is selected, the output-path noise gate is only activated (i.e., muted) when all of the respective signal thresholds are satisfied.



For example, if the OUT1L noise gate is associated with the OUT1L and OUT1R signal paths, the OUT1L signal path is only muted if both the OUT1L and OUT1R signal levels are below the respective thresholds.

The noise-gate threshold (the signal level below which the noise gate is activated) is set using NGATE_THR. Note that, for each output path, the noise-gate threshold represents the signal level at the respective output pins; the threshold is therefore independent of the digital volume and PGA gain settings.

Note that, although there is only one noise-gate threshold level (NGATE_THR), each output-path noise gate may be activated independently, according to the respective signal content and the associated threshold configurations.

To prevent erroneous triggering, a time delay is applied before the gate is activated; the noise gate is only activated (i.e., muted) when the output levels are below the applicable signal level thresholds for longer than the noise-gate hold time. The hold time is set using the NGATE_HOLD field.

When the noise gate is activated, the CS42L92 gradually attenuates the respective signal path at the rate set by OUT_VD_RAMP (see Table 4-77). When the noise gate is deactivated, the output volume increases at the rate set by OUT_VI_RAMP.

Register Address	Bit	Label	Default	Description				
R1043 (0x0413)	11:0	OUT1L_NGATE_	0x001	Output Signal Path Noise-Gate Source. Enables one of more signal paths as				
Noise_Gate_Select_1L		SRC[11:0]		inputs to the respective noise gate. If more than one signal path is enabled as				
R1047 (0x0417)	11:0	OUT1R_NGATE_	0x002	an input, the noise gate is only activated (i.e., muted) when all of the respective signal thresholds are satisfied.				
Noise_Gate_Select_1R		SRC[11:0]		Each bit is coded as 0 = Disabled, 1 = Enabled				
R1051 (0x041B)	11:0	OUT2L_NGATE_	0x004	[11] = Reserved				
Noise_Gate_Select_2L		SRC[11:0]		[10] = Reserved				
R1055 (0x041F)	11:0	OUT2R_NGATE_	0x008	[9] = OUT5R				
Noise_Gate_Select_2R		SRC[11:0]		[8] = OUT5L				
R1059 (0x0423)	11:0	OUT3L_NGATE_	0x010	[0] = Reserved				
Noise_Gate_Select_3L		SRC[11:0]		[6] = Reserved				
R1063 (0x0427)	11:0	OUT3R_NGATE_	0x020	[5] = OUT3R				
Noise_Gate_Select_3R		SRC[11:0]		[4] = OUT3L				
R1075 (0x0433)	11:0	OUT5L_NGATE_	0x100	[3] = OUT2R				
Noise_Gate_Select_5L		SRC[11:0]		[2] = OUT2L				
R1079 (0x0437)	11:0	OUT5R_NGATE_	0x200	[1] = OUT1R				
Noise_Gate_Select_5R		SRC[11:0]		[0] = OUT1L				
R1112 (0x0458)	5:4	NGATE	00	Output Signal Path Noise-Gate Hold Time (delay before noise gate is activated)				
Noise Gate Control		HOLD[1:0]		00 = 30 ms 10 = 250 ms				
				01 = 120 ms 11 = 500 ms				
	3:1	NGATE_THR[2:0]	000	Output Signal Path Noise-Gate Threshold				
				000 = -78 dB 011 = -96 dB 110 = -114 dB				
				001 = -84 dB 100 = -102 dB 111 = -120 dB				
				010 = -90 dB 101 = -108 dB				
	0	NGATE_ENA	0	Output Signal Path Noise-Gate Enable				
				0 = Disabled				
				1 = Enabled				

Table 4-79. Output Signal Path Noise-Gate Control

4.11.7 Output Signal Path AEC Loop-Back

The CS42L92 incorporates two loop-back signal paths, which are ideally suited as a reference for AEC processing. Any two of the output signal paths may be selected as the AEC loop-back sources.

When configured with suitable DSP firmware, the CS42L92 can provide an integrated AEC capability. The AEC loop-back feature also enables convenient hook-up to an external device for implementing the required signal-processing algorithms.

The AEC loop-back source is connected after the respective digital volume controls, as shown in Fig. 4-52. The AEC loop-back signals can be selected as input to any of the digital mixers within the CS42L92 digital core. The sample rate for the AEC loop-back paths is configured using OUT_RATE—see Table 4-26.



The AEC loop-back function is enabled using the AEC*n*_LOOPBACK_ENA bits (where *n* identifies the applicable path, AEC1 or AEC2). The source signals for the Transmit Path AEC function are selected using the AEC*n*_LOOPBACK_SRC bits.

The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the AEC loop-back function. If the frequency is too low, an attempt to enable this function fails. Note that active signal paths are not affected under such circumstances.

The AEC*n*_ENA_STS bits indicate the status of the AEC loop-back functions. If an underclocked error condition occurs, these bits indicate whether the AEC loop-back function has been enabled.

Register Address	Bit	Label	Default	Description
R1104 (0x0450)	5:2	AEC1_LOOPBACK_	0000	Input source for Tx AEC1 function
DAC_AEC_		SRC[3:0]		0000 = OUT1L 0100 = OUT3L All other codes are reserved
Control_1				0001 = OUT1R 0101 = OUT3R
				0010 = OUT2L 1000 = OUT5L
				0011 = OUT2R 1001 = OUT5R
	1	AEC1_ENA_STS	0	Transmit (Tx) Path AEC1 Control Status
				0 = Disabled
				1 = Enabled
	0	AEC1_LOOPBACK_	0	Transmit (Tx) Path AEC1 Control
		ENA		0 = Disabled
				1 = Enabled
R1105 (0x0451)	5:2	AEC2_LOOPBACK_	0000	Input source for Tx AEC2 function
DAC_AEC_		SRC[3:0]		0000 = OUT1L 0100 = OUT3L All other codes are reserved
Control_2				0001 = OUT1R 0101 = OUT3R
				0010 = OUT2L 1000 = OUT5L
				0011 = OUT2R 1001 = OUT5R
	1	AEC2_ENA_STS	0	Transmit (Tx) Path AEC2 Control Status
				0 = Disabled
				1 = Enabled
	0	AEC2_LOOPBACK_	0	Transmit (Tx) Path AEC2 Control
		ENA		0 = Disabled
				1 = Enabled

Table 4-80. Output Signal Path AEC Loop-Back Control

4.11.8 Headphone Outputs

The headphone/earpiece driver outputs, HPOUT1–HPOUT4, are suitable for direct connection to external headphones and earpieces. The outputs are ground referenced, eliminating any requirement for AC coupling capacitors.

The headphone outputs incorporate a common mode, or ground loop, feedback path that provides rejection of system-related ground noise. The feedback connection is configurable for each output path. Five feedback pins are provided (HPOUTFB*n*). The selected feedback pins must be connected to ground for normal operation of the headphone outputs.

Note that the feedback pins should be connected to GND as close as possible to the respective headphone jack ground pin, as shown in Fig. 4-54. In mono (differential) mode, the feedback pins should be connected to the ground plane that is closest to the earpiece output PCB tracks.

The ground feedback path for HPOUT1 and HPOUT2 headphone outputs is selected using the HP1_GND_SEL and HP2_ GND_SEL register fields respectively—see Table 4-81. The ground feedback path for HPOUT3 and HPOUT4 headphone outputs is selected using the HP3_GND_SEL field.

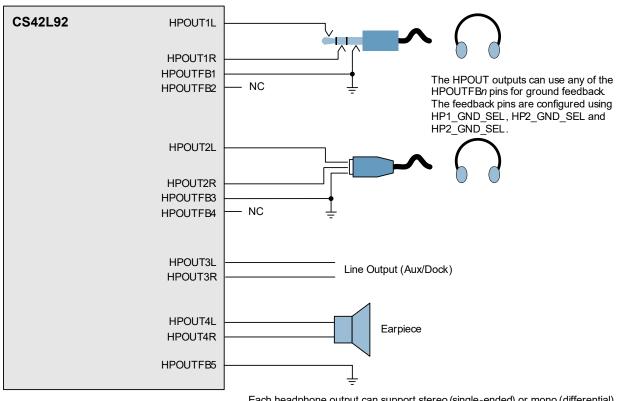
If none of the HPOUTFB*n* pins are available for use as a ground feedback connection, an internal ground (AGND) can be selected by setting the respective HP*n*_GND_SEL field to 101.



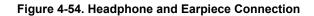
Register Address	Bit	Label	Default	Description
R1042 (0x0412)	2:0	HP1_GND_	000	HPOUT1 ground feedback pin select
Output_Path_		SEL[2:0]		000 = HPOUTFB1
Config_1				001 = HPOUTFB2
				010 = HPOUTFB3
				011 = HPOUTFB4
				100 = HPOUTFB5
				101 = AGND (internal connection)
				All other codes are reserved
R1050 (0x041A)	2:0	HP2_GND_	010	HPOUT2 ground feedback pin select
Output_Path_		SEL[2:0]		000 = HPOUTFB1
Config_2				001 = HPOUTFB2
				010 = HPOUTFB3
				011 = HPOUTFB4
				100 = HPOUTFB5
				101 = AGND (internal connection)
				All other codes are reserved
R1058 (0x0422)	2:0	HP3_GND_	010	HPOUT3/HPOUT4 ground feedback pin select
Output_Path_		SEL[2:0]		000 = HPOUTFB1
Config_3				001 = HPOUTFB2
				010 = HPOUTFB3
				011 = HPOUTFB4
				100 = HPOUTFB5
				101 = AGND (internal connection)
				All other codes are reserved

Table 4-81. Headphone Output (HPOUT) Ground Feedback Control

The headphone and earpiece connections are shown in Fig. 4-54.



Each headphone output can support stereo (single-ended) or mono (differential) output. The illustration shows the configuration for a typical application .





4.11.9 Speaker Outputs (Digital PDM)

The CS42L92 supports a two-channel pulse-density modulation (PDM) digital speaker interface; the PDM outputs are associated with the OUT5L and OUT5R output signal paths.

The external connections associated with the PDM outputs are implemented on multifunction GPIO pins, which must be configured for the respective PDM functions when required. The PDM output connections are pin-specific alternative functions available on specific GPIO pins. See Section 4.14 to configure the GPIO pins for the PDM output.

The PDM digital speaker interface is a stereo interface; the OUT5L and OUT5R output signal paths are interleaved on the SPKDAT output, and clocked using SPKCLK.

Note that the PDM interface supports two different operating modes; these are selected using SPK1_FMT. See Table 3-15 for detailed timing information in both modes.

- If SPK1_FMT = 0 (Mode A), the Left PDM channel is valid at the rising edge of SPKCLK; the Right PDM channel is
 valid at the falling edge of SPKCLK.
- If SPK1_FMT = 1 (Mode B), the Left PDM channel is valid during the low phase of SPKCLK; the Right PDM channel is valid during the high phase of SPKCLK.

The PDM interface timing is shown in Fig. 4-55.

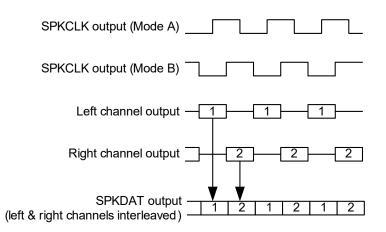


Figure 4-55. Digital Speaker (PDM) Interface Timing

Clocking for the PDM interface is derived from SYSCLK. Note that SYSCLK_ENA must also be set. See Section 4.16 for further details of the system clocks and control registers.

If the OUT5L or OUT5R output signal path is enabled, the PDM interface clock signal is output on the SPKCLK pin.

The output signal paths support normal and high performance operating modes, as described in Section 4.11.3. The SPKCLK frequency is set according to the operating mode of the relevant output path, as described in Table 4-82. The OUT5_OSR bit is defined in Table 4-74.

Note that the SPKCLK frequencies noted in Table 4-82 assume that the SYSCLK frequency is a multiple of 6.144 MHz (SYSCLK_FRAC = 0). If the SYSCLK frequency is a multiple of 5.6448 MHz (SYSCLK_FRAC=1), the SPKCLK frequency is scaled accordingly.

Table 4-82. SPKCLK Frequency

OUT5_OSR	Description	SPKCLK Frequency
0	Normal mode	3.072 MHz
1	High Performance mode	6.144 MHz

The PDM output channels can be independently muted. When muted, the default output on each channel is a DSD-compliant silent stream (0110_1001b). The mute output code can be programmed to other values if required, using the SPK1_MUTE_SEQ field. The mute output code can be transmitted MSB-first or LSB-first; this is selectable using the SPK1_MUTE_ENDIAN bit.



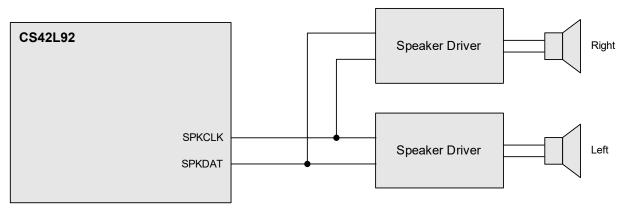
Note that the PDM Mute function is not a soft-mute; the audio output is interrupted immediately when the PDM mute is asserted. It is recommended to use the output signal path mute function before applying the PDM mute. See Table 4-77 for details of the OUT5L_MUTE and OUT5R_MUTE bits.

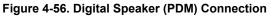
The PDM output interface registers are described in Table 4-83.

Register Address	Bit	Label	Default	Description
R1168 (0x0490)	13	SPK1R_MUTE	0	PDM Speaker Output 1 (Right) Mute
PDM_SPK1_				0 = Audio output (OUT5R)
CTRL_1				1 = Mute Sequence output
	12	SPK1L_MUTE	0	PDM Speaker Output 1 (Left) Mute
				0 = Audio output (OUT5L)
				1 = Mute Sequence output
	8	SPK1_MUTE_	0	PDM Speaker Output 1 Mute Sequence Control
		ENDIAN		0 = Mute sequence is LSB first
				1 = Mute sequence output is MSB first
	7:0	SPK1_MUTE_	0x69	PDM Speaker Output 1 Mute Sequence
		SEQ[7:0]		Defines the 8-bit code that is output on SPKDAT (left) or SPKDAT (right) when muted.
R1169 (0x0491)	0	SPK1_FMT	0	PDM Speaker Output 1 timing format
PDM_SPK1_				0 = Mode A (PDM data is valid at the rising/falling edges of SPKCLK)
CTRL_2				1 = Mode B (PDM data is valid during the high/low phase of SPKCLK)

Table 4-83. Digital Speaker (PDM) Output Control

The digital speaker (PDM) outputs SPKDAT and SPKCLK are intended for direct connection to a compatible external speaker driver. A typical configuration is shown in Fig. 4-56.





4.12 External Accessory Detection

The CS42L92 provides external accessory detection functions that can sense the presence and impedance of external components. This can be used to detect the insertion or removal of an external headphone or headset, and to provide an indication of key/button push events.

Jack insertion is detected using the JACKDETn pins (where n = 1, 2, or 3), which are typically connected to a switch contact within the jack sockets. An interrupt event is generated whenever a jack insertion or jack removal event is detected.

Suppression of pops and clicks caused by jack insertion or removal is provided using the MICDET clamp function. (The CS42L92 incorporates two MICDET clamps.) This function can also be used to trigger interrupt events, and to trigger the control-write sequencer. The integrated general-purpose switches can be synchronized with the MICDET clamps, to provide additional pop-suppression capability.

Microphones, push buttons and other accessories can be detected via the MICDET*n* pins. The presence of a microphone, and the status of a hook switch can be detected. This feature can also be used to detect push-button operation. (Note that accessory detection is also possible via the HPDET*n* and JACKDET*n* pins, subject to some additional constraints.)



Headphone impedance can be measured via the HPDET1 and HPDET2 pins; this can be used to set different gain levels or other configuration settings according to the type of load connected. For example, different settings may be applicable to headphone or line output loads. (Note that impedance measurement is also possible via the MICDET*n* and JACKDET*n* pins, subject to some additional constraints.)

The MICVDD power domain must be enabled when using the microphone detect function. (Note that MICVDD is not required for the jack detect or headphone detect functions.) The MICVDD power domain is provided using an internal charge pump (CP2) and LDO regulator (LDO2). See Section 4.19 for details of these circuits.

The internal 32-kHz clock must be present and enabled when using the microphone detect or headphone detect functions; the 32-kHz clock is also required for the jack detect function, assuming input debounce is enabled. See Section 4.16 for details of the internal 32-kHz clock and associated control fields.

4.12.1 Jack Detect

The CS42L92 provides support for jack insertion switch detection. The jack insertion status can be read using the relevant register status bits. A jack insertion or removal can also be used to trigger an interrupt event.

The jack-detect interrupt (IRQ) functionality is maintained in Sleep Mode (see Section 4.13). This enables a jack insertion event to be used to trigger a wake-up of the CS42L92.

Jack insertion and removal is detected using the JACKDET*n* pins. The recommended external connections are shown in Fig. 4-57. Note that the logic thresholds associated with the JACKDET*n* pins differ from each other, as described in Table 3-11—this provides support for different jack switch configurations.

The jack detect feature is enabled using the JD n_ENA bits (where n = 1, 2, or 3 for JACKDET1, JACKDET2, or JACKDET3 respectively); the jack insertion status can be read using JD n_STSx . Note that the JD n_STS1 and JD n_STS2 bits provide the same information in respect of the applicable JACKDETn input.

The jack detect input debounce is selected using the JD*n*_DB bits, as described in Table 4-84. Note that, under normal operating conditions, the debounce circuit uses the 32-kHz clock, which must be enabled whenever input debounce functions are required. Input debounce is not provided in Sleep Mode; the JD*n*_DB bits have no effect in Sleep Mode.

Note that the jack detect signals (JD*n*) can be used as inputs to the MICDET clamp function—this provides additional functionality relating to jack insertion and removal events.

An interrupt request (IRQ) event is generated whenever a jack insertion or jack removal is detected (see Section 4.15). Separate mask bits are provided, to allow IRQ events on the rising and/or falling edges of the JD*n* signals.

The control registers associated with the jack detect function are described in Table 4-84.

Register Address	Bit	Label	Default	Description
R723 (0x02D3)	2	JD3_ENA	0	JACKDET3 enable
Jack_detect_				0 = Disabled
analog				1 = Enabled
	1	JD2_ENA	0	JACKDET2 enable
				0 = Disabled
				1 = Enabled
	0	JD1_ENA	0	JACKDET1 enable
				0 = Disabled
				1 = Enabled

Table 4-84. Jack Detect Control



Register Address	Bit	Label	Default	Description
R6278 (0x1886)	8	JD3_STS1	0	JACKDET3 input status
IRQ1_Raw_				0 = Jack not detected
Status_7				1 = Jack is detected
				(Assumes the JACKDET3 pin is pulled low on jack insertion.)
	2	JD2_STS1	0	JACKDET2 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET2 pin is pulled low on jack insertion.)
	0	JD1_STS1	0	JACKDET1 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET1 pin is pulled low on jack insertion.)
R6534 (0x1986)	8	JD3_STS2	0	JACKDET3 input status
IRQ2_Raw_				0 = Jack not detected
Status_7				1 = Jack is detected
				(Assumes the JACKDET3 pin is pulled low on jack insertion.)
	2	JD2_STS2	0	JACKDET2 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET2 pin is pulled low on jack insertion.)
	0	JD1_STS2	0	JACKDET1 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET1 pin is pulled low on jack insertion.)
R6662 (0x1A06)	8	JD3_DB	0	JACKDET3 input debounce
Interrupt_				0 = Disabled
Debounce_7				1 = Enabled
	2	JD2_DB	0	JACKDET2 input debounce
				0 = Disabled
				1 = Enabled
	0	JD1_DB	0	JACKDET1 input debounce
				0 = Disabled
				1 = Enabled

Table 4-84. Jack Detect Control (Cont.)

A recommended connection circuit, including headphone output on HPOUT1 and microphone connections, is shown in Fig. 4-57. See Section 5.1 for details of recommended external components.



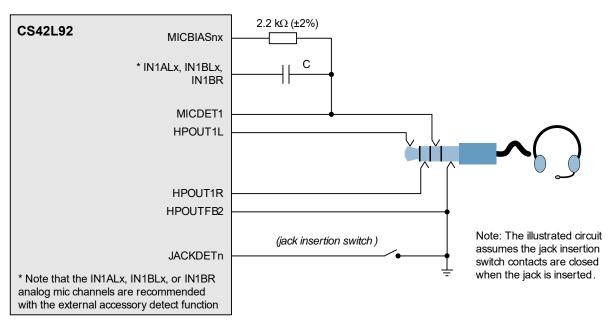
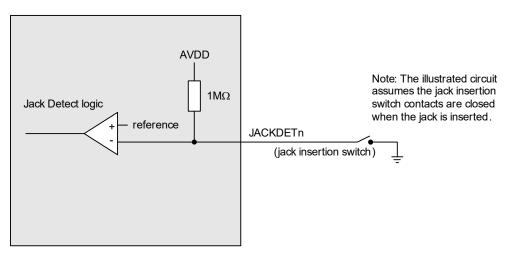


Figure 4-57. Jack Detect and External Accessory Connections

The internal comparator circuit used to detect the JACKDET*n* status is shown in Fig. 4-58. The threshold voltages for the jack detect circuit are noted in Table 3-11. Note that separate thresholds are defined for jack insertion and removal.





4.12.2 Jack Pop Suppression (MICDET Clamp and GP Switch)

Under typical configuration of a 3.5-mm headphone/accessory jack connection, there is a risk of pops and clicks arising from jack insertion or removal. This can occur if the headphone load makes momentary contact with the MICBIAS output when the jack is not fully inserted.

The CS42L92 provides a MICDET clamp function to suppress pops and clicks caused by jack insertion or removal. Two MICDET clamps are incorporated; each clamp can be controlled directly, or can be activated by a logic function derived from the JACKDET*n* inputs. The clamp status can be read using the relevant register status bits. The clamp status can also be used to trigger an interrupt (IRQ) event or to trigger the control-write sequencer.

Two general-purpose analog switches are incorporated; these can be configured to augment the MICDET clamp functions and used to support the pop-suppression circuits, as described in Section 4.12.2.3.



Note that, due to control logic that is shared between the two MICDET clamps, some restrictions regarding the operation of the clamps must be observed, as described in Section 4.12.2.1.

4.12.2.1 MICDET Clamp Control

The MICDET clamp function can be configured using the MICD_CLAMPn_MODE fields (where n = 1 or 2 for Clamp 1 or Clamp 2 respectively). Selectable logic conditions (derived from the JD1, JD2, or JD3 signals—see Table 4-84) provide support for different jack detect circuit configurations. Setting the MICD_CLAMPn_OVD bit enables the respective MICDET clamp, regardless of other conditions.

Note: The MICD_CLAMP*n*_OVD bits are set by default. Accordingly, the MICDET clamps are always enabled following power-on reset, hardware reset, or software reset.

The MICDET clamp functionality (including the external IRQ) is maintained in Sleep Mode (see Section 4.13). This enables a jack insertion event to be used to trigger a wake-up of the CS42L92. The recommended control sequence for the jack detect and MICDET clamp control is described in Section 4.12.2.5.

The MICDET clamps are effective on pins MICDET1–MICDET4, as follows:

- If MICDET Clamp 1 is enabled, the MICDET1/HPOUTFB1 and MICDET2/HPOUTFB2 pins are shorted together.
- If MICDET Clamp 2 is enabled, the MICDET3/HPOUTFB3 and MICDET4/HPOUTFB4 pins are shorted together.

If either clamp is enabled, the grounding of the applicable MICDET is achieved via the HPOUTFB function of the other pin associated with the respective clamp. It is assumed that the HPOUTFB connection is grounded externally, as shown in Fig. 4-59.

The selectable logic conditions supported by the MICD_CLAMP*n*_MODE fields provide flexibility in selecting the appropriate conditions for controlling each MICDET clamp. The MICD_CLAMP1_MODE field allows Clamp 1 to be controlled by the JD1 or JD2 signals. The MICD_CLAMP2_MODE field allows Clamp 2 to be controlled by the JD3 signal.

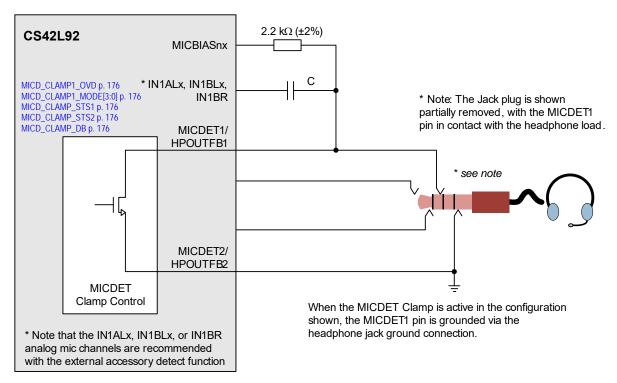
Note that, due to control logic that is shared between the two clamps, the option to control both clamps in response to the JD*n* signals cannot be supported at the same time. If automatic (JACKDET-triggered) operation is selected on one of the clamps, the other clamp must be held in a fixed state. It is recommended that the unused clamp should be enabled by setting the respective MICD_CLAMP*n*_OVD bit.

The status of the clamps can be read using the MICD_CLAMP_STSx bits. Note that the MICD_CLAMP_STS1 and MICD_ CLAMP_STS2 bits provide the same information. The MICD_CLAMP_STSx bits indicate the status of the active clamp only (i.e., the clamp that is not in the overridden state). The status of a clamp in the overridden (MICD_CLAMP $_$ OVD = 1) state is not indicated. It is assumed that a maximum of one clamp is active at any time.

The MICDET clamp debounce is selected by setting MICD_CLAMP_DB bit, as described in Table 4-85. Note that, under normal operating conditions, the debounce circuit uses the 32-kHz clock, which must be enabled whenever input debounce functions are required. Input debounce is not provided in Sleep Mode; the MICD_CLAMP_DB bit has no effect in Sleep Mode.

The MICDET clamp function is shown in Fig. 4-59. Note that the jack plug is shown partially removed, with the MICDET1 pin in contact with the headphone load.







4.12.2.2 Interrupts and Write-Sequencer Control

An interrupt request (IRQ) event can be generated in response to the MICDET clamp status. A MICDET clamp interrupt is generated whenever the logic condition of the JD*n* signals cause a change in the clamp status. Separate maskable interrupts are provided for the rising and falling edges of the MICDET clamp status—see Section 4.15.

The control-write sequencer can be triggered by the MICDET clamp status. This is enabled using the WSEQ_ENA_MICD_ CLAMP_FALL and WSEQ_ENA_MICD_CLAMP_RISE bits. Note that these control-sequencer events are only valid if the clamp status changed in response to the JD*n* signals. See Section 4.18 for details of the control-write sequencer.

Note that, due to control logic that is shared between the two MICDET clamps, the option to control both clamps in response to the JD*n* signals cannot be supported at the same time. It is assumed that a maximum of one clamp is active at any time. Accordingly, only the active clamp is capable of generating an interrupt event or triggering the control-write sequencer.

4.12.2.3 Pop Suppression using General-Purpose Switch

In applications where a large decoupling capacitance is present on the MICBIAS output, the MICDET clamp function may be unable to discharge the capacitor sufficiently to eliminate pops and clicks associated with jack insertion and removal. In this case, it may be desirable to use one of the general-purpose switches on the CS42L92 to provide isolation from the MICBIAS output.

There are two general-purpose switches, configured using the respective SWn_MODE fields (where n = 1 or 2). The SWn_MODE fields allow the switches to be disabled, enabled, or synchronized to the MICDET clamp status, as described in Table 4-85.

For jack pop suppression, it is recommended to set $SWn_MODE = 11$ for the applicable switch. In this case, the switch contacts are open whenever the MICDET clamp status bits are set (clamp enabled), and the switch contacts are closed whenever the MICDET clamp status bits are clear (clamp disabled).

Note that the MICDET clamp status (MICD_CLAMP_STSx) is shared between the two clamps. Under recommended operating conditions, a maximum of one clamp should be active at any time (see Section 4.12.2.1). In this case, the MICD_CLAMP_STSx bits indicate the status of the active clamp only.



A typical pop-suppression circuit, incorporating the general-purpose switch and MICDET clamp function, is shown in Fig. 4-60. Normal accessory functions are supported when the switch contacts (GPSW*n*P and GPSW*n*N) are closed, and the MICDET clamp is disabled. Ground clamping of MICDET, and isolation of MICBIAS, are achieved when the switch contacts are open and the MICDET clamp is enabled.

Note that the MICDET clamp function must also be configured appropriately if using this method of pop-suppression control.

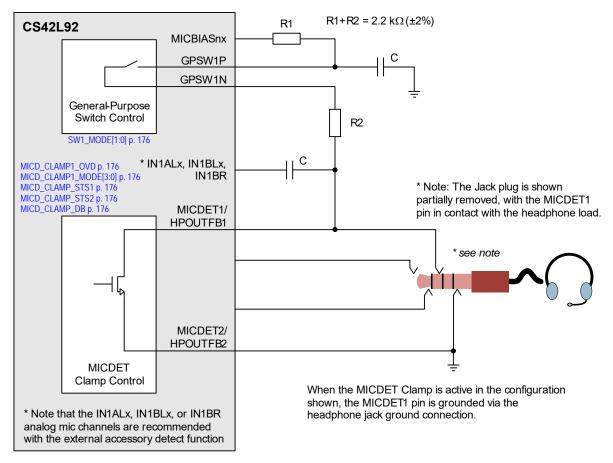


Figure 4-60. General-Purpose Switch Circuit

4.12.2.4 MICDET Clamp Control Registers

The control registers associated with the MICDET clamp and general-purpose switch functions are described in Table 4-85.

Register Address	Bit	Label	Default	Description
R65 (0x0041)	7	WSEQ_ENA_	0	MICDET Clamp (Falling) Write Sequencer Select
Sequence_control		MICD_CLAMP_		0 = Disabled
		FALL		1 = Enabled
	6	WSEQ_ENA_	0	MICDET Clamp (Rising) Write Sequencer Select
	MICD_CLAMP_		0 = Disabled	
		RISE		1 = Enabled

 Table 4-85. MICDET Clamp and General-Purpose Switch Control



Register Address	Bit	Label	Default	Description
R710 (0x02C6)	9	MICD_	1	MICDET Clamp 2 Override
Micd_Clamp_		CLAMP2_OVD		0 = Disabled (clamp is controlled by MICD_CLAMP2_MODE)
control				1 = Enabled (clamp is enabled)
	8:6	MICD_	000	MICDET Clamp 2 Mode
		CLAMP2_		0x0 = Disabled
		MODE[2:0]		0x1 = Enabled (MICDET3/MICDET4 shorted together)
				0x2 = Enabled if JD3=0
				0x3 = Enabled if JD3=1
	4	MICD_	1	MICDET Clamp 1Override
		CLAMP1_OVD		0 = Disabled (clamp is controlled by MICD_CLAMP1_MODE)
				1 = Enabled (clamp is enabled)
	3:0	MICD_	0000	MICDET Clamp 1 Mode
		CLAMP1_		0x0 = Disabled 0x9 = Enabled if JD1=0 or JD2=1
		MODE[3:0]		0x1 = Enabled (MICDET1/MICDET2 shorted together) 0xA = Enabled if JD1=1 or JD2=0
				0x2–0x3 = Reserved 0xB = Enabled if JD1=1 or JD2=1
				0x4 = Enabled if JD1=0 0xC = Enabled if JD1=0 and JD2=0
				0x5 = Enabled if JD1=1 0xD = Enabled if JD1=0 and JD2=1
				0x6 = Enabled if JD2=0 0xE = Enabled if JD1=1 and JD2=0
				0x7 = Enabled if JD2=1 0xF = Enabled if JD1=1 and JD2=1
				0x8 = Enabled if JD1=0 or JD2=0
R712 (0x02C8)	3:2	SW2_	00	General-purpose Switch 2 control
GP_Switch_1		MODE[1:0]		00 = Disabled (switch open) 10 = Enabled if MICDET clamp status is set
				01 = Enabled (switch closed) 11 = Enabled if MICDET clamp status is clear
	1:0	SW1_	00	General-purpose Switch 1 control
		MODE[1:0]		00 = Disabled (switch open) 10 = Enabled if MICDET clamp status is set
				01 = Enabled (switch closed) 11 = Enabled if MICDET clamp status is clear
R6278 (0x1886)	4	MICD_CLAMP_	0	MICDET Clamp status
IRQ1_Raw_Status_		STS1		0 = Clamp disabled
7				1 = Clamp enabled
				Note: Separate _STS bits are not provided for each clamp—it is assumed that a
				maximum of one clamp is active at any time. The clamp override condition (MICD_ CLAMP n OVD = 1) is not indicated.
R6534 (0x1986)	4	MICD CLAMP	0	MICDET Clamp status
IRQ2_Raw_Status_	+	STS2	0	0 = Clamp disabled
7				1 = Clamp enabled
-				Note: Separate STS bits are not provided for each clamp—it is assumed that a
				maximum of one clamp is active at any time. The clamp override condition (MICD
				CLAMPn_OVD = 1) is not indicated.
R6662 (0x1A06)	4	MICD_CLAMP_	0	MICDET Clamp debounce
Interrupt_		DB		0 = Disabled
Debounce 7			1	1 = Enabled

Table 4-85. MICDET Clamp and General-Purpose Switch Control (Cont.)

4.12.2.5 Control Sequence for Jack Detect and MICDET Clamp

A summary of the jack detect and MICDET clamp functionality, and the recommended usage in typical applications, is described as follows:

 On device power-up, and following reset, the MICDET clamps are enabled due to the default setting of the MICD_ CLAMPn_OVD bits; this ensures no spurious output can occur during jack insertion. It is recommended to keep the MICDET clamps enabled (MICD_CLAMPn_OVD = 1) until after a jack insertion has been detected.

To control MICDET Clamp 1 according to the JD1/JD2 signals, the MICDET_CLAMP1_MODE field should be set according to the required logic condition (configured to enable the clamp when jack is removed).

To control MICDET Clamp 2 according to the JD3 signal, the MICDET_CLAMP2_MODE field should be set according to the required logic condition (configured to enable the clamp when jack is removed).

Note that, due to control logic that is shared between the two MICDET clamps, the option to control both clamps in response to the JD*n* signals cannot be supported at the same time. It is assumed that a maximum of one clamp is active at any time.

 Jack insertion is indicated using the JDn signals or MICDET clamp interrupts (assuming that the MICD_CLAMPn_ MODE field has been correctly set for the applicable JDn signal configuration); the associated status bits can be read directly, or associated signals can be unmasked as inputs to the interrupt controller.

After jack insertion has been detected, the applicable headset functions (headphone, microphone, accessory detect) may then be enabled.

If the headset function requires MICBIAS to be enabled on the respective jack, the associated MICDET clamp should be disabled (MICD_CLAMP $n_OVD = 0$) immediately before enabling the MICBIAS (or immediately before enabling MICD n_ENA). Note that, if MICBIAS is not required on the respective jack, the clamp should not be disabled (e.g., for headphone-only operation).

• Jack removal is also indicated using the JD*n* signals or MICDET clamp interrupts. The associated status bits can be read directly, or can be unmasked as inputs to the interrupt controller. The MICDET clamp ensures fast and automatic silencing of the jack outputs.

Under typical use cases, the respective MICBIAS generator and headset audio paths should all be disabled following jack removal.

After jack removal has been detected, the MICDET clamp override bit (MICD_CLAMP*n*_OVD) should be set, to make the system ready for a jack insertion.

The recommended control sequence for jack detect and MICDET clamp is summarized in Table 4-86.

Event	Device Actions	Recommended User Actions		
Initial condition	Clamp enabled by default	Configure MICD_CLAMPn_MODE		
Jack insertion	Jack insertion signaled via IRQ	For headphone-only operation:		
		Enable output signal paths		
		For other use cases:		
		Disable clamp, MICD_CLAMP <i>n</i> _OVD = 0		
		Enable MICBIAS and MICDET		
		Enable I/O signal paths		
Jack removal	Jack removal signaled via IRQ,	Disable MICBIAS and MICDET		
	Clamp enabled automatically	Disable I/O signal paths		
		Enable clamp MICD_CLAMP <i>n</i> _OVD = 1		

Table 4-86. Control Sequence for Jack Detect and MICDET Clamp

4.12.3 Microphone Detect

The CS42L92 microphone detection circuit measures the impedance of an external load connected to one of the MICDET pins. This feature can be used to detect the presence of a microphone, and the status of the associated hook switch. It can also be used to detect push-button status or the connection of other external accessories.

4.12.3.1 Microphone Detect Control

The microphone detection circuit measures the external impedance connected to the MICDET*n* pins. In the discrete measurement mode, the function reports whether the measured impedance lies within one of eight predefined levels. In the ADC measurement mode, a more specific result is provided in the form of a 7-bit ADC output.

Note that microphone/accessory detection is also possible via the HPDET*n* and JACKDET*n* pins, subject to some additional constraints. If the measurement (sense) pin is connected to MICVDD or MICBIAS*nx* (typically via a 2.2-k Ω bias resistor), MICDET*n* must always be used.

The microphone detection circuit typically uses one of the MICBIAS outputs as a reference. The CS42L92 automatically enables the appropriate MICBIAS output when required in order to perform the detection function; this allows the detection function to be supported in low-power standby operating conditions.

The MICVDD power domain must be enabled when using the microphone detection function. This power domain is provided using an internal charge pump (CP2) and LDO regulator (LDO2). See Section 4.19 for details of these circuits. The internal 32-kHz clock must be present and enabled when using the microphone detection function; see Section 4.16 for details.



The CS42L92 provides two microphone detection circuits, which are independently configurable. Detection can be enabled on both circuits simultaneously.

To configure the microphone detection circuit, the applicable pin connections for the intended measurement must be written to the $MICDn_SENSE_SEL$ and $MICDn_GND_SEL$ fields (where *n* identifies the respective detection circuit, 1 or 2). The respective detection circuit measures the external impedance between the pins selected by these two fields; the valid selections for each are defined in Table 4-87.

Note: There is no requirement for the SENSE and GND pin selections to be uniquely assigned between the microphone detect and headphone detect functions—the same pin may be used as a SENSE or GND connection for more than one of the detection functions. If multiple microphone/headphone detections are enabled, the respective measurements are automatically scheduled in isolation to each other. See Section 4.12.4 for details of the headphone detect function.

The microphone detection circuit uses MICVDD, or any one of the MICBIAS*xy* sources, as a reference. The applicable source is configured using the MICD*n*_BIAS_SRC field. If HPDET*n* or JACKDET*n* is selected as the measurement pin, MICD*n*_BIAS_SRC should be set to 1111.

The microphone detection function is enabled by setting MICDn_ENA.

When microphone detection is enabled, the CS42L92 performs a number of measurements in order to determine the external impedance between the selected pins. The measurement process is repeated at a cyclic rate controlled by MICD*n*_RATE. The MICD*n*_RATE field selects the delay between completion of one measurement and the start of the next. When the microphone detection result has settled, the CS42L92 indicates valid data by setting MICD*n*_VALID.

The discrete measurement mode and ADC measurement mode provide different capabilities for microphone detection. The control requirements and the measurement indication mechanisms differ according to the selected mode, as follows:

 In the discrete measurement mode (MICDn_ADC_MODE = 0), the measured impedance is only deemed valid after more than one successive measurement has produced the same result. The MICDn_DBTIME field provides control of the debounce period; this can be either two measurements or four measurements.

When the microphone detection result has settled (i.e., after the applicable debounce period), the CS42L92 indicates valid data by setting the MICD n_VALID bit. The measured impedance is indicated using the MICD n_LVL and MICD n_STS bits, as described in Table 4-87.

The MICDn_VALID bit, when set, remains asserted for as long as the microphone detection function is enabled (i.e., while MICDn_ENA = 1). If the detected impedance changes, the MICDn_LVL and MICDn_STS fields change, but the MICDn_VALID bit remains set, indicating valid data at all times.

The detection circuit supports up to eight impedance levels (including the no-accessory-detected level), enabling detection of a typical microphone and up to six push buttons. Each measurement level can be enabled or disabled independently; this provides flexibility according to the required thresholds, and offers a faster measurement time in some applications. The MICD*n*_LVL_SEL field is described in Section 4.12.3.3. The default configuration supports a maximum of four push buttons, in accordance with the AndroidTM wired headset specification.

Note that, for typical headset detection, the choice of external resistance values must take into account the impedance of the microphone—the detected impedance corresponds to the combined parallel resistance of the microphone and any asserted push button. Examples of suitable external components are described in Section 5.1.7.

In the ADC measurement mode (MICDn_ADC_MODE = 1), the detection function generates two output results, contained within the MICDn_ADCVAL and MICDn_ADCVAL_DIFF fields. These fields contain the most recent measurement value (MICDn_ADCVAL) and the measurement difference value (MICDn_ADCVAL_DIFF). The difference value indicates the difference between the latest measurement and the previous measurement; this can be used to determine whether the measurement is stable and reliable.

In ADC measurement mode, the detection function must be disabled before the measurement can be read. When the CS42L92 indicates valid data (MICD $n_VALID = 1$), the detection must be disabled by setting MICD $n_ENA = 0$. Note that MICD n_ADCVAL and MICD n_ADCVAL_DIFF do not follow a linear coding. The appropriate test condition

for accepting the measurement value (or for rescheduling the measurement) varies depending on the application requirements, and depending on the expected impedance value.



The microphone detection functions are inputs to the interrupt control circuit and can be used to trigger an interrupt event every time an accessory insertion, removal, or impedance change is detected; see Section 4.15.

The fields associated with microphone detection (or other accessories) are described in Table 4-87. The external circuit configuration is shown in Fig. 4-61.

Register Address	Bit	Label	Default	Description				
R674 (0x02A2)	15	MICD1_ADC_	0	Mic Detect 1 Measurement	Mode			
Mic_Detect_1_		MODE		0 = Discrete Mode				
Control_0				1 = ADC Mode				
	7:4	MICD1_SENSE_	0001	Mic Detect 1 Sense Select				
		SEL[3:0]		0000 = MICDET1	0100 = HPDET1	1000 = MICDET5		
				0001 = MICDET2	0101 = HPDET2	1001 = JACKDET3		
				0010 = MICDET3	0110 = JACKDET1	All other codes are		
				0011 = MICDET4	0111 = JACKDET2	reserved		
	2:0							
		SEL[2:0]		000 = MICDET1/HPOUTFE	31 011 = MICI	DET4/HPOUTFB4		
				001 = MICDET2/HPOUTFE	32 100 = MICI	DET5/HPOUTFB5		
				010 = MICDET3/HPOUTFE	33 All other co	odes are reserved		
R675 (0x02A3) Mic Detect 1	15:12	MICD1_BIAS_ STARTTIME[3:0]	0001	Mic Detect 1 Bias Start-up Delay (Selects the delay time between enabling the MICBIASnx reference and performing the MICDET function.)				
Control 1				0000 = 0 ms (continuous)		1010 = 128 ms		
_				0001 = 0.25 ms	0110 = 8 ms	1011 = 256 ms		
				0010 = 0.5 ms	0111 = 16 ms	1100 = 512 ms		
				0011 = 1 ms	1000 = 32 ms	1101 = 24 ms		
				0100 = 2 ms	1001 = 64 ms	1110 to 1111 = 512 ms		
	11:8	MICD1_	0001	Mic Detect 1 Rate (Selects the delay between successive MICDET measurements.)				
		RATE[3:0]		0000 = 0 ms (continuous)	0101 = 4 ms	1010 = 128 ms		
				0001 = 0.25 ms	0110 = 8 ms	1011 = 256 ms		
				0010 = 0.5 ms	0111 = 16 ms	1100 = 512 ms		
				0011 = 1 ms	1000 = 32 ms	1101 = 24 ms		
				0100 = 2 ms	1001 = 64 ms	1110 to 1111 = 512 ms		
	7:4							
		SRC[3:0]		0000 = MICBIAS1A	0011 = MICBIAS1D	1111 = MICVDD		
				0001 = MICBIAS1B	0100 = MICBIAS2A	All other codes are		
				0010 = MICBIAS1C	0101 = MICBIAS2B	reserved		
	1	MICD1_DBTIME	1	Mic Detect 1 Debounce				
				0 = 2 measurements				
				1 = 4 measurements				
				Only valid when MICD1_ADC_MODE = 0.				
	0	MICD1_ENA 0 Mic Detect 1 Enable						
				0 = Disabled				
				1 = Enabled				
R676 (0x02A4)	7:0	MICD1_LVL_	1001_			on in specific impedance ranges)		
Mic_Detect_1_		SEL[7:0]	1111	[7] = Enable 1–30 k Ω detection [3] = Not used				
Control_2				[6] = Not used		e 360–680 Ω detection		
				[5] = Not used	[1] = Enabl	e 210–290 Ω detection		
				[4] = Not used	[0] = Enabl	e 110–180 Ω detection		
				Only valid when MICD1_A	$DC_MODE = 0.$			

Table 4-87. Microphone Detect Control



Register Address	Bit	Label	Default		Description		
R677 (0x02A5)	10:2	MICD1_LVL[8:0]		Mic Detect 1 Level (indicates the measured impedance)			
Mic_Detect_1_			0000	[8] = 1–30 kΩ	[3] = 360–	-680 Ω	
Control_3				[7] = Not used	[2] = 210–	-290 Ω	
				[6] = Not used	[1] = 110–	180 Ω	
				[5] = Not used	[0] = 0–70	Ω	
				[4] = Not used			
					uding loads >30 k Ω , may a	npedance limits. Note that other also be indicated using these	
	1	MICD1_VALID	0	Mic Detect 1 Data Valid			
				0 = Not Valid			
				1 = Valid			
	0	MICD1_STS	0	Mic Detect 1 Status			
				0 = Mic/accessory not dete	ected		
				1 = Mic/accessory detected			
				Mic/accessory detection is	assured for load impedan	ce up to 30 kΩ.	
	Only valid when MICD1_ADC_MODE = 0.						
R683 (0x02AB)	15:8	MICD1_	0x00	Mic Detect 1 ADC Level (Difference)			
Mic_Detect_1_ Control 4		ADCVAL_ Only valid when MICD1_ADC_MODE = 1.					
	6:0	MICD1_	0x00	Mic Detect 1 ADC Level			
		ADCVAL[6:0]		Only valid when MICD1_ADC_MODE = 1.			
R690 (0x02B2)	15				t Mode		
Mic_Detect_2_		MODE		0 = Discrete Mode			
Control_0				1 = ADC Mode			
	7:4	MICD2_SENSE_	0001	Mic Detect 2 Sense Select			
		SEL[3:0]		0000 = MICDET1	0100 = HPDET1	1000 = MICDET5	
				0001 = MICDET2	0101 = HPDET2	1001 = JACKDET3	
				0010 = MICDET3	0110 = JACKDET1	All other codes are	
				0011 = MICDET4	0111 = JACKDET2	reserved	
	2:0 MICD2_GND_ SEL[2:0]		000	Mic Detect 2 Ground Selec			
				000 = MICDET1/HPOUTFI			
				001 = MICDET2/HPOUTFI		DET5/HPOUTFB5	
				010 = MICDET3/HPOUTFI	B3 All other c	odes are reserved	

Table 4-87. Microphone Detect Control (Cont.)



Mic_Detect_2_		MICD2 BIAS	0001	Mic Detect 2 Bias Start up	Delay (Calenta the delay time	
			0001	Mic Detect 2 Bias Start-up Delay (Selects the delay time between enabling the		
		STARTTIME[3:0]			performing the MICDET fun	
Control_1				0000 = 0 ms (continuous)		1010 = 128 ms
				0001 = 0.25 ms	0110 = 8 ms	1011 = 256 ms
				0010 = 0.5 ms	0111 = 16 ms	1100 = 512 ms
				0011 = 1 ms	1000 = 32 ms	1101 = 24 ms
				0100 = 2 ms	1001 = 64 ms	1110 to 1111 = 512 ms
1		MICD2_	0001	•	•	ive MICDET measurements.)
		RATE[3:0]		0000 = 0 ms (continuous)	0101 = 4 ms	1010 = 128 ms
				0001 = 0.25 ms	0110 = 8 ms	1011 = 256 ms
				0010 = 0.5 ms	0111 = 16 ms	1100 = 512 ms
				0011 = 1 ms	1000 = 32 ms	1101 = 24 ms
				0100 = 2 ms	1001 = 64 ms	1110 to 1111 = 512 ms
-		MICD2_BIAS_	0000	Mic Detect 2 Reference Sel		
		SRC[3:0]		0000 = MICBIAS1A	0011 = MICBIAS1D	1111 = MICVDD
				0001 = MICBIAS1B	0100 = MICBIAS2A	All other codes are
				0010 = MICBIAS1C	0101 = MICBIAS2B	reserved
	1	MICD2_DBTIME	1	Mic Detect 2 Debounce		
				0 = 2 measurements		
				1 = 4 measurements		
				Only valid when MICD2_AI	DC_MODE = 0.	
	0	MICD2_ENA	0	Mic Detect 2 Enable		
				0 = Disabled		
				1 = Enabled		
R692 (0x02B4)		MICD2_LVL_	1001_			n in specific impedance ranges)
Mic_Detect_2_		SEL[7:0]	1111	[7] = Enable 1–30 k Ω detection	ction [3] = Not use	ed
Control_2				[6] = Not used	[2] = Enable	360–680 Ω detection
				[5] = Not used	[1] = Enable	210–290 Ω detection
				[4] = Not used	[0] = Enable	110–180 Ω detection
				Only valid when MICD2_AI		
	10:2	MICD2_LVL[8:0]	0_0000_		es the measured impedance	,
Mic_Detect_2_			0000	[8] = 1–30 kΩ	[3] = 360–68	
Control_3				[7] = Not used	[2] = 210–29	
				[6] = Not used	[1] = 110–18	
				[5] = Not used	[0] = 0–70 Ω	2
				[4] = Not used		
						edance limits. Note that other
				impedance conditions, inclubits. Only valid when MICD	uding loads >30 k Ω , may als	o be indicated using these
	1		0	Mic Detect 2 Data Valid	$IZ_ADC_IMODE = 0.$	
	1	MICD2_VALID	0	0 = Not Valid		
–	0			1 = Valid		
	0	MICD2_STS	0	Mic Detect 2 Status	atad	
				0 = Mic/accessory not dete		
				1 = Mic/accessory detected		
					assured for load impedance	e up to 30 K22.
	15.0	MICDO	000	Only valid when MICD2_AI		
. ,		MICD2_ ADCVAL	0x00	Mic Detect 2 ADC Level (D		
Mic_Detect_2_		DIFF[7:0]		Only valid when MICD2_AI	$DC_MODE = 1.$	
Control 4			1			
Control_4		MICD2	0x00	Mic Detect 2 ADC Level		

Table 4-87. Microphone Detect Control (Cont.)

The external connections for the microphone detect circuit are shown in Fig. 4-61. In typical applications, it can be used to detect a microphone or button press.

Note that, when using the microphone detect circuit, it is recommended to use the IN1AL*x*, IN1BL*x*, or IN1BR analog microphone input paths to ensure best immunity to electrical transients arising from the external accessory.

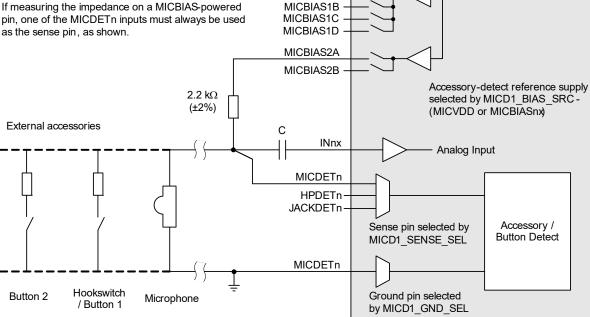


I DO2

Regulator

Note that the IN1ALx, IN1BLx, or IN1BR analog mic channels are recommended for use with the external accessory detect function.

If measuring the impedance on a MICBIAS-powered pin, one of the MICDETn inputs must always be used as the sense pin, as shown.



MICVDD

MICBIAS1A

Figure 4-61. Microphone- and Accessory-Detect Interface

MICBIAS Reference Control 4.12.3.2

The voltage reference for the microphone detection is configured using the MICDn BIAS SRC field, as described in Table 4-87. The microphone detection function automatically enables the applicable reference when required for impedance measurement.

If the selected reference (MICBIASxy) is not already enabled, the microphone detect circuit automatically enables the respective MICBIAS output for short periods of time only, every time the impedance measurement is scheduled. To allow time for the associated circuitry to stabilize, a time delay is applied before the measurement is performed; this is configured using MICDn BIAS STARTTIME, as described in Table 4-87. If the measurement rate setting (MICDn RATE) is greater than 0x0, the delay (MICDn BIAS STARTTIME) should be set to 0.25 ms or more.

The microphone detection automatically enables the applicable MICBIASxy output switch, every time the Note: impedance measurement is scheduled. The respective MICBIAS generator (MICBIAS1 or MICBIAS2) is not controlled automatically-the applicable generators must be enabled using the MICB1 ENA and MICB2 ENA bits, as described in Table 4-124.

The timing of the microphone detect function is shown in Fig. 4-62. Two different cases are shown, according to whether MICBIASxy is enabled periodically by the impedance measurement function, or is enabled at all times.

If the selected reference (MICBIASxy) is not enabled continuously, the respective MICBIASxy discharge bits should be cleared. The MICBIAS control registers are described in Section 4.19.



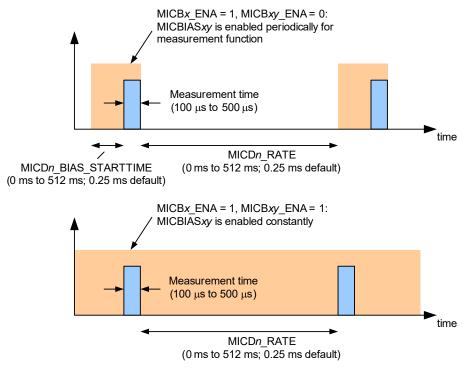


Figure 4-62. Microphone- and Accessory-Detect Timing

4.12.3.3 Measurement Range Control

When the discrete measurement mode is selected (MICD $n_ADC_MODE = 0$), the MICD $n_LVL_SEL[7:0]$ bits allow each impedance measurement level to be enabled or disabled independently. This allows the function to be tailored to the particular application requirements.

If one or more bits within MICD*n*_LVL_SEL is cleared, the corresponding impedance level is disabled. Any measured impedance which lies in a disabled level is reported as the next lowest, enabled level.

For example, the MICD*n*_LVL_SEL[2] bit enables the detection of a 360–680 Ω impedance. If MICD*n*_LVL_SEL[2] = 0, an external impedance in this range is indicated in the next lowest detection range (210–290 Ω); this would be reported in the MICD*n*_LVL field as MICD*n*_LVL[2] = 1.

With default register configuration, and all measurement levels enabled, the CS42L92 can detect the presence of a typical microphone and up to four push buttons. It is possible to configure the detection circuit for up to eight push buttons, by adjusting the impedance detection thresholds. However, adjustment of the detection thresholds is outside the scope of this data sheet—please contact your local Cirrus Logic representative for further information, if required.

The measurement time varies between 100–500 μ s, depending on the impedance of the external load, and depending on how many impedance measurement levels are enabled. A high impedance is measured faster than a low impedance.

4.12.3.4 External Components

The external connections for the microphone detect circuit are shown in Fig. 4-61. Examples of suitable external components are described in Section 5.1.7.

The accuracy of the microphone detect function is assured whenever the connected load is within the applicable limits specified in Table 3-11. It is required that a 2.2-k Ω (2%) resistor must also be connected between the measurement (SENSE) pin and the selected MICBIAS reference—different resistor values lead to inaccuracy in the impedance measurement.

Note that, for typical headset detection, the choice of external resistance values must take into account the impedance of the microphone—the detected impedance corresponds to the combined parallel resistance of the microphone and any asserted push button.



4.12.4 Headphone Detect

The CS42L92 headphone detection circuit measures the impedance of an external headphone load. This feature can be used to set different gain levels or to apply other configuration settings according to the type of load connected. Separate monitor pins are provided for headphone detection on the left and right channels of HPOUT*n*.

4.12.4.1 Headphone Detection Control

The headphone detection circuit measures the external impedance connected to the HPDET*n* pins. In typical usage, this provides measurement of the load impedance on one or more of the headphone outputs (HPOUT1–4).

Note that impedance measurement is also possible via the MICDET*n* and JACKDET*n* pins, subject to some additional constraints. If the measurement (sense) pin is connected to one of the headphone outputs, then HPDET1, HPDET2, or JACKDET1 must always be used. The valid measurement range and the measurement accuracy are reduced, if using the MICDET*n* or JACKDET*n* pins.

To configure the headphone detection circuit, the applicable pin connections for the intended measurement must be written to the HPD_SENSE_SEL and HPD_GND_SEL fields. The headphone detection circuit measures the external impedance between the pins selected by these two fields; the valid selections for each are defined in Table 4-90.

When measuring the load impedance on one the HPOUT*n* output paths, the HPD_GND_SEL selection should be the same MICDET*n*/HPOUTFB*n* pin as the ground feedback pin for the applicable headphone output. See Section 4.11.8 to configure the ground feedback pin for the HPOUT*n* outputs.

The HPD_FRC_SEL field must also be configured, to select where the measurement current is applied. As a general rule, this should be the same as the HPD_SENSE_SEL pin. Other configurations can be used if required—for example, to improve measurement accuracy in cases where the SENSE input path includes significant unwanted resistance.

Note: There is no requirement for the SENSE and GND pin selections to be uniquely assigned between the microphone detect and headphone detect functions—the same pin may be used as a SENSE or GND connection for more than one of the detection functions. If multiple microphone/headphone detections are enabled, the respective measurements are automatically scheduled in isolation to each other. See Section 4.12.3 for details of the microphone detect function.

Headphone detection is commanded by writing 1 to HPD_POLL.

The impedance measurement range is configured using HPD_IMPEDANCE_RANGE. This field should be set in accordance with the expected load impedance. Note that a number of separate measurements are typically required to determine the load impedance; the recommended control requirements are described in Section 4.12.4.2.

Note: Setting HPD_IMPEDANCE_RANGE is not required for detection on the MICDET*n* or JACKDET*n* pins. The impedance measurement range, and measurement accuracy, in these cases are different to the HPDET1 and HPDET2 measurements.

If headphone detection is performed using a measurement pin connected to one of the headphone outputs, the respective output driver must be disabled before the measurement is commanded. The required settings are shown in Table 4-88.

Description	Requirement
HPOUT1L Impedance measurement	HPD_OVD_ENA = 1, HPD_OUT_SEL = 000, HP1L_ENA = 0
HPOUT1R Impedance measurement	HPD_OVD_ENA = 1, HPD_OUT_SEL = 001, HP1R_ENA = 0
HPOUT2L Impedance measurement	HPD_OVD_ENA = 1, HPD_OUT_SEL = 010, HP2L_ENA = 0
HPOUT2R Impedance measurement	HPD_OVD_ENA = 1, HPD_OUT_SEL = 011, HP2R_ENA = 0
HPOUT3L or HPOUT4L Impedance measurement	HPD_OVD_ENA = 1, HPD_OUT_SEL = 100, HP3L_ENA = 0
HPOUT3R or HPOUT4R Impedance measurement	HPD_OVD_ENA = 1, HPD_OUT_SEL = 101, HP3R_ENA = 0

Table 4-88. Output Configuration for Headphone Detect

Note: The applicable headphone outputs configuration must be maintained until after the headphone detection has completed. See Table 4-71 for details of the HP*nx*_ENA bits.

If headphone detection is performed using a measurement pin that is not connected to one of the headphone outputs, the HPD_OVD_ENA bit should be cleared.



If headphone detection is performed using a measurement pin that is also connected to one of the MICBIAS outputs, the respective MICBIAS output must be disabled and floating (MICBnx_ENA = 0, MICBnx_DISCH = 0).

When headphone detection is commanded, the CS42L92 uses an adjustable current source to determine the connected impedance. A sweep of measurement currents is applied. The rate of this sweep can be adjusted using HPD_CLK_DIV and HPD_RATE.

4.12.4.2 Measurement Output

The headphone detection process typically comprises a number of separate measurements (for different impedance ranges). Completion of each measurement is indicated by HPD_DONE. When this bit is set, the measurement result can be read from the HPD_DACVAL field, and decoded as described in Eq. 4-3.

Impedance (
$$\Omega$$
) =
$$\frac{C_0 + (C_1 \times Offset)}{\left[\frac{(HPD_DACVAL + 0.5)}{C_2}\right] - \left[\frac{1}{C_3(1 + (C_4 \times Gradient))}\right]} - C_5$$

Equation 4-3. Headphone Impedance Calculation

The associated parameters for decoding the measurement result are defined Table 4-89. The applicable values are dependent on the HPD_IMPEDANCE_RANGE setting in each case. The *Offset* and *Gradient* values are derived from register fields that are factory-calibrated for each device.

Parameter	HPD_IMPEDANCE_ RANGE = 00	HPD_IMPEDANCE_ RANGE = 01	HPD_IMPEDANCE_ RANGE = 10	HPD_IMPEDANCE_ RANGE = 11
C ₀	1.007	1.007	9.744	100.684
C ₁	-0.0072	-0.0072	-0.0795	-0.9494
C ₂	4005	7975	7300	7300
C ₃	69.3	69.6	62.9	63.2
C ₄	0.0055	0.0055	0.0055	0.0055
C ₅	0.6	0.6	0.6	0.6
Offset	HP_OFFSET_00	HP_OFFSET_01	HP_OFFSET_10	HP_OFFSET_11
Gradient	HP_GRADIENT_0X	HP_GRADIENT_0X	HP_GRADIENT_1X	HP_GRADIENT_1X

Table 4-89. Headphone Measurement Decode Parameters

Note that, to achieve the specified measurement accuracy, the above equation must be calculated to an accuracy of at least 5 decimal places throughout.

The impedance measurement result is valid if $169 \le HPD_DACVAL \le 1019$. (In case of any contradiction with the HPD_IMPEDANCE_RANGE description, the HPD_DACVAL validity takes precedence.)

If the external impedance is entirely unknown (i.e., it could lie in any of the HPD_IMPEDANCE_RANGE regions), it is recommended to test initially with HPD_IMPEDANCE_RANGE = 00. If the resultant HPD_DACVAL is < 169, the impedance is higher than the selected measurement range, so the test should be scheduled again, after incrementing HPD_IMPEDANCE_RANGE.

Each measurement is triggered by writing 1 to HPD_POLL. Completion of each measurement is indicated by HPD_DONE. Note that, after HPD_DONE bit has been asserted, it remains asserted until the next measurement has been commanded.

Note: A simpler, but less accurate, procedure for headphone impedance measurement is also supported, using the HPD_LVL field. When the HPD_DONE bit is set, indicating completion of a measurement, the impedance can be read directly from the HPD_LVL field, provided that the value lies within the range of the applicable HPD_IMPEDANCE_RANGE setting.

Note that, for detection using the MICDET*n* or JACKDET*n* pins, the HPD_LVL field is the only supported measurement output option. The HPD_IMPEDANCE_RANGE field is not valid for detection on the MICDET*n* or JACKDET*n* pins. See Table 4-90 for further description of the HPD_LVL field.

The headphone detection function is an input to the interrupt control circuit and can be used to trigger an interrupt event on completion of the headphone detection; see Section 4.15.



The fields associated with headphone detection are described in Table 4-90. The external circuit configuration is shown Fig. 4-63.

Note that 32-bit register addressing is used from R12888 (0x3000) upwards; 16-bit format is used otherwise. The registers noted in Table 4-90 contain a mixture of 16- and 32-bit register addresses.

Register Address	Bit	Label	Default	De	scription	
R665 (0x0299)	15	HPD_OVD_ENA	0	Headphone Detect Output Overr	ide Enable	
Headphone_ Detect_0				This bit, when set, causes the HPD_OUT_SEL headphone output channel to be automatically configured for headphone detection ea time headphone detection is scheduled. Note that the respective ou driver must also be disabled (HPnx_ENA = 0) for the duration of a headphone output impedance measurement. 0 = Disabled 1 = Enabled		
	14:12	HPD_OUT_SEL[2:0]	000	Headphone Detect Output Chan	nel Select	
				000 = HPOUT1L	100 = HPOUT3L	
				001 = HPOUT1R	101 = HPOUT3R	
				010 = HPOUT2L	All other codes are reserved	
				011 = HPOUT2R		
	11:8	HPD_FRC_SEL[3:0]	0000	Headphone Detect Measuremen	t Current Pin Select	
				0000 = MICDET1	0110 = JACKDET1	
				0001 = MICDET2	0111 = JACKDET2	
				0010 = MICDET3	1000 = MICDET5	
				0011 = MICDET4	1001 = JACKDET3	
				0100 = HPDET1	All other codes are reserved	
				0101 = HPDET2		
	7:4	HPD_SENSE_SEL[3:0]	0000	Headphone Detect Sense Pin Se		
				0000 = MICDET1	0110 = JACKDET1	
				0001 = MICDET2	0111 = JACKDET2	
				0010 = MICDET3	1000 = MICDET5	
				0011 = MICDET4	1001 = JACKDET3	
				0100 = HPDET1	All other codes are reserved	
					0101 = HPDET2	
	2:0	HPD_GND_SEL[2:0]	000	Headphone Detect Ground Pin S		
				000 = MICDET1/HPOUTFB1	011 = MICDET4/HPOUTFB4	
				001 = MICDET2/HPOUTFB2	100 = MICDET5/HPOUTFB5	
				010 = MICDET3/HPOUTFB3	All other codes are reserved	

Table 4-90. Headphone Detect Control



Register Address	Bit	Label	Default	Description
R667 (0x029B)	10:9	HPD_IMPEDANCE_	00	Headphone Detect Range
Headphone_		RANGE[1:0]		$00 = 4 \Omega$ to 30Ω
Detect_1				01 = 8 Ω to 100 Ω
				$10 = 100 \Omega$ to $1 k\Omega$
				11 = 1 kΩ to 10 kΩ
				Only valid when HPD_SENSE_SEL = 0100 or 0101.
	4:3	HPD_CLK_DIV[1:0]	00	Headphone Detect Clock Rate (Selects the clocking rate of the headphone detect adjustable current source. Decreasing the clock rate gives a slower measurement time.) 00 = 32 kHz 01 = 16 kHz
				10 = 8 kHz
	0.1		00	11 = 4 kHz
	2:1	HPD_RATE[1:0]	00	Headphone Detect Sweep Rate (Selects the step size between successive measurements. Increasing the step size gives a faster measurement time.)
				00 = 1
				01 = 2
				10 = 4
				11 = Reserved
	0	HPD_POLL	0	Headphone Detect Enable
				Write 1 to start HP Detect function
R668 (0x029C)	15	HPD_DONE	0	Headphone Detect Status
Headphone_				0 = HP Detect not complete
Detect_2				1 = HP Detect done
	14:0	HPD_LVL[14:0]	0x0000	Headphone Detect Level
				$LSB = 0.5 \Omega$
				$8 = 4 \Omega \text{ or less}$
				9 = 4.5 Ω
				10 = 5 Ω
				11 = 5.5 Ω
				 20,000 = 10 kΩ or more
				For HPDET1 or HPDET2 measurement (HPD_SENSE_SEL = 0100 or 0101), HPD_LVL is valid from 4 Ω to10 k Ω , within the range selected by HPD_IMPEDANCE_RANGE.
				For other measurements, HPD_LVL is valid from 400 Ω to 6 k Ω only.
				If HPD_LVL reports a value outside the valid range, the range should be adjusted and the measurement repeated. A $0-\Omega$ result may be reported if the measurement is less than the minimum value for the selected range.
R669 (0x029D)	9:0	HPD_DACVAL[9:0]	0x000	Headphone Detect Level (Coded as integer, LSB = 1).
Headphone_ Detect_3				See separate description for full decode information.

Table 4-90. Headphone Detect Control (Cont.)



Register Address	Bit	Label	Default	Description
R131076	31:24	HP_OFFSET_11[7:0]	See	Headphone Detect Calibration field.
(0x20004)			Footnote 1	Signed number, LSB = 0.25.
OTP_HPDET_Cal_				Range is –31.75 to +31.75.
1				Default value is factory-set per device.
	23:16	HP_OFFSET_10[7:0]	See	Headphone Detect Calibration field.
			Footnote 1	Signed number, LSB = 0.25.
				Range is –31.75 to +31.75.
				Default value is factory-set per device.
	15:8	HP_OFFSET_01[7:0]	See	Headphone Detect Calibration field.
			Footnote 1	Signed number, LSB = 0.25.
				Range is –31.75 to +31.75.
				Default value is factory-set per device.
	7:0	HP_OFFSET_00[7:0]	See	Headphone Detect Calibration field.
			Footnote 1	Signed number, LSB = 0.25.
				Range is –31.75 to +31.75.
				Default value is factory-set per device.
R131078	15:8	HP_GRADIENT_1X[7:0]	See	Headphone Detect Calibration field.
(0x20006)			Footnote 1	Signed number, LSB = 0.25.
OTP_HPDET_Cal_				Range is –31.75 to +31.75.
2				Default value is factory-set per device.
	7:0	HP_GRADIENT_0X[7:0]	See	Headphone Detect Calibration field.
			Footnote 1	Signed number, LSB = 0.25.
				Range is –31.75 to +31.75.
				Default value is factory-set per device.

Table 4-90. Headphone Detect Control (Cont.)

1. Default value is factory-set per device.

The external connections for the headphone detect circuit are shown in Fig. 4-63.

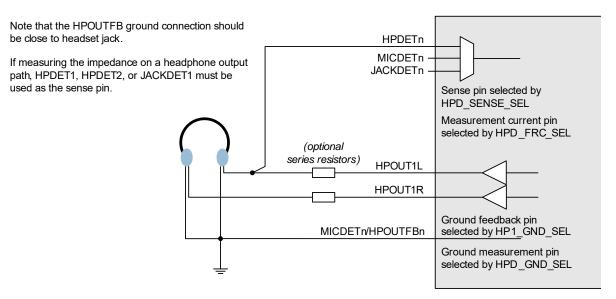


Figure 4-63. Headphone Detect Interface

Note that, where external resistors are connected in series with the headphone load, as shown, it is recommended that the HPDET*n* connection is to the headphone side of the resistors. If the HPDET*n* connection is made to the CS42L92 end of these resistors, this leads to a corresponding offset in the measured impedance.

Under default conditions, the measurement time varies between 17–244 ms, depending on the impedance of the external load. A high impedance is measured faster than a low impedance.



4.13 Low Power Sleep Configuration

The CS42L92 supports a low-power Sleep Mode, in which most functions are disabled and power consumption is minimized. The CS42L92 enters Sleep Mode when the DCVDD supply is removed. Note that the AVDD and DBVDD supplies must be present throughout the Sleep Mode duration.

In Sleep Mode, the CS42L92 can generate an interrupt event in response to a change in voltage on the JACKDET1, JACKDET2, or JACKDET3 pins. This enables a jack insertion event (or other digital logic transition) to be used to trigger a wake-up of the CS42L92.

In Sleep Mode, the CS42L92 can provide an unregulated voltage output on the MICBIAS1A pin. This can be used to power an external microphone during Sleep Mode—see Section 4.19.

The system clocks (SYSCLK, ASYNCCLK, DSPCLK) should be disabled before selecting Sleep Mode. The external clock input (MCLK*n*) may also be stopped, if desired.

The functionality and control fields associated with Sleep Mode are supported via an internal always-on supply domain. The always-on control registers are listed in Table 4-91. These fields are maintained (i.e., not reset) in Sleep Mode.

Note that the control interface is not supported in Sleep Mode; read/write access to the always-on registers is not possible. Access to the register map using any of the control interfaces should be ceased before selecting Sleep Mode.

Register Address	Label	Reference
R710 (0x02C6)	MICD_CLAMP2_OVD	See Section 4.12
	MICD_CLAMP2_MODE[2:0]	
	MICD_CLAMP1_OVD	
	MICD_CLAMP1_MODE[3:0]	
R723 (0x02D3)	MICB1A_AOD_ENA	See Section 4.19
	JD3_ENA	See Section 4.12
	JD2_ENA	
	JD1_ENA	
R6150 (0x1806)	MICD_CLAMP2_FALL_EINT1	See Section 4.15
	MICD_CLAMP2_RISE_EINT1	
	JD3_FALL_EINT1	
	JD3_RISE_EINT1	
	MICD_CLAMP1_FALL_EINT1	
	MICD_CLAMP1_RISE_EINT1	
	JD2_FALL_EINT1	
	JD2_RISE_EINT1	
	JD1_FALL_EINT1	
	JD1_RISE_EINT1	
R6214 (0x1846)	IM_MICD_CLAMP2_FALL_EINT1	
	IM_MICD_CLAMP2_RISE_EINT1	
	IM_JD3_FALL_EINT1	
	IM_JD3_RISE_EINT1	
	IM_MICD_CLAMP1_FALL_EINT1	
	IM_MICD_CLAMP1_RISE_EINT1	
	IM_JD2_FALL_EINT1	
	IM_JD2_RISE_EINT1	
	IM_JD1_FALL_EINT1	
	IM_JD1_RISE_EINT1	1
R6784 (0x1A80)	IM_IRQ1	1
	IRQ_POL	1
	IRQ_OP_CFG	1
R6864 (0x1AD0)	RESET_PU	See Section 4.23
	RESET_PD	1

Table 4-91. Sleep Mode Always-On Control Registers

The always-on digital I/O pins are listed in Table 4-92. All other digital input pins have no effect in Sleep Mode; all other digital output pins are undriven (floating).



Note: The IN*nx*/DMIC*x* connections are isolated from the CS42L92 circuits in Sleep Mode. This enables a microphone that is connected to the CS42L92 to be used by another circuit while Sleep Mode is selected.

The IRQ output is normally deasserted in Sleep Mode. In Sleep Mode, the IRQ output can be asserted only in response to the JACKDET1, JACKDET2, or JACKDET3 inputs. If the IRQ output is asserted in Sleep Mode, it can be deasserted only after a wake-up transition.

Output drivers and bus keepers are disabled in Sleep Mode, for all pins not on the always-on domain; this means that the logic level on these pins is undefined. If a defined logic state is required during Sleep Mode (e.g., as input to another device), an external pull resistor may be required. If an external pull resistor is connected to a pin that also supports a bus keeper function, the pull resistance should be chosen carefully, taking into account the resistance of the bus keeper. See Section 4.14.1 for specific notes concerning the GPIO pins.

Pin Name	Description	Reference
IRQ	Interrupt Request output	See Section 4.15
JACKDET1	Jack Detect input 1	See Section 4.12
JACKDET2	Jack Detect input 2	
JACKDET3	Jack Detect input 3	
RESET	Digital Reset input (active low)	See Section 4.23

Table 4-92. Sleep Mode Always-On Digital Input/Output Pins

The always-on functionality includes the JD1, JD2, and JD3 control signals, which provide support for the low-power Sleep Mode. The MICDET clamp status signal is also supported; this is controlled by a selectable logic function, derived from JD1, JD2, or JD3.

The JD1, JD2, JD3, and MICDET clamp status signals are derived from the JACKDET1, JACKDET2, and JACKDET3 inputs, and can be used to trigger the interrupt controller.

- The JD1, JD2, and JD3 signals are derived from the jack detect function (see Section 4.12). These inputs can be used to trigger a response to a jack insertion or jack removal detection.
 If these signals are enabled, the JD1, JD2, and JD3 signals indicate the status of the JACKDET1, JACKDET2, and JACKDET3 input pins respectively. See Table 4-84 for details of the associated control fields.
- The MICDET clamp status is controlled by the JD1, JD2, or JD3 signals (see Section 4.12). The configurable logic provides flexibility in selecting the appropriate conditions for activating the MICDET clamp. The clamp status can be used to trigger a response to a jack insertion or jack removal detection.

The MICDET clamp function is configured using MICD_CLAMP1_MODE and MICD_CLAMP2_MODE, as described in Table 4-85. Note that, due to control logic that is shared between the two clamps, the option to control both clamps in response to the JD*n* signals cannot be supported at the same time. It is assumed that a maximum of one clamp is active at any time—the MICDET clamp status provides an indication for the active clamp only.

The interrupt functionality associated with these signals is part of the always-on functionality, enabling the CS42L92 to provide indication of jack insertion or jack removal to the host processor in Sleep Mode; see Section 4.15.

Note that the JACKDET*n* inputs do not result in a wake-up transition directly; a wake-up transition only occurs by reapplication of DCVDD. In a typical application, the JACKDET*n* inputs provide a signal to the applications processor, via the IRQ output; if a wake-up transition is required, this is triggered by the applications processor enabling the DCVDD supply.

4.14 General-Purpose I/O

The CS42L92 supports up to 16 GPIO pins, which can be assigned to application-specific functions. The GPIOs enable interfacing and detection of external hardware and can provide logic outputs to other devices. The GPIO input functions can be used to generate an interrupt (IRQ) event.



There are 2 dedicated GPIO pins; the remaining 14 GPIOs are implemented as alternate functions to a pin-specific capability. The GPIO and interrupt circuits support the following functions:

- Pin-specific alternative functions for external interfaces (AIF, PDM)
- Logic input/button detect (GPIO input)
- Logic 1 and Logic 0 output (GPIO output)
- Interrupt (IRQ) status output
- Clock output
- Frequency-locked loop (FLL) status output
- FLL clock output
- IEC-60958-3-compatible S/PDIF output
- Pulse-width modulation (PWM) signal output
- ASRC lock status
- General-purpose timer status output
- Event logger FIFO buffer status output
- · Alarm generator status output
- Auxiliary PDM interface

Logic input and output (GPIO) can be supported in two different ways on the CS42L92. The standard mechanism described in this section provides a comprehensive suite of options including input debounce, and selectable output drive configuration. The DSP GPIO circuit is tailored towards more advanced requirements typically demanded by DSP software features. The DSP GPIO functions are described in Section 4.5.4.

The CS42L92 also incorporates two general-purpose switches; these are analog switches, described in Section 4.14.18.

If the JTAG interface is enabled, the GPIO13–15 pins are configured as a JTAG interface that provides test and debug access to the CS42L92. The respective GPIO configuration registers have no effect in this case, and the GPIO pins cannot be assigned any other function. See Section 4.20 for details of the JTAG interface.

4.14.1 GPIO Control

For each GPIO, the selected function is determined by the GP n_FN field, where n identifies the GPIO pin (1–16). The pin direction, set by GP n_DIR , must be set according to function selected by GP n_FN .

If a pin is configured as a GPIO input ($GPn_DIR = 1$, $GPn_FN = 0x001$), the logic level at the pin can be read from the respective GPn_LVL bit. Note that GPn_LVL is not affected by the GPn_POL bit.

A debounce circuit can be enabled on any GPIO input, to avoid false event triggers. This is enabled on each pin by setting the respective GP*n*_DB bit. The debounce circuit uses the 32-kHz clock, which must be enabled whenever input debounce functions are required. The debounce time is configurable using the GP_DBTIME field. See Section 4.16 for further details of the CS42L92 clocking configuration.

Each GPIO pin is an input to the interrupt control circuit and can be used to trigger an interrupt event. An interrupt event is triggered on the rising and falling edges of the GPIO input. The associated interrupt bit is latched once set; it can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.

Integrated pull-up and pull-down resistors are provided on each GPIO pin; these can be configured independently using the GP*n*_PU and GP*n*_PD fields. When the pull-up and pull-down control bits are both enabled, the CS42L92 provides a bus keeper function on the respective pin. The bus keeper function holds the logic level unchanged whenever the pin is undriven (e.g., if the signal is tristated).

Note: The bus keeper is enabled by default on all GPIO pins and, if not actively driven, may result in either a Logic 0 or Logic 1 at the respective input on start-up. If an external pull resistor is connected (e.g., to control the logic level in Sleep Mode), the chosen resistance should take account of the bus keeper resistance (see Table 3-10). A strong pull resistor (e.g., 10 kΩ) is required, if a specific start-up condition is to be forced by the external pull component.



If a pin is configured as a GPIO output ($GPn_DIR = 0$, $GPn_FN = 0x001$), its level can be set to Logic 0 or Logic 1 using the GPn_LVL field. Note that the GPn_LVL bits are write-only when the respective GPIO pin is configured as an output.

If a pin is configured as an output ($GPn_DIR = 0$), the polarity can be inverted using the GPn_POL bit. When $GPn_POL = 1$, the selected output function is inverted. In the case of logic level output ($GPn_FN = 0x001$), the external output is the opposite logic level to GPn_LVL when $GPn_POL = 1$. Note that, if $GPn_FN = 0x000$ or 0x002, the GPn_POL bit has no effect on the respective GPIO pin.

A GPIO output can be either CMOS driven or open drain. This is selected on each pin using the respective GPn_OP_CFG bit. Note that if $GPn_FN = 0x000$ the GPn_OP_CFG bit has no effect on the respective GPIO pin—see Table 4-93 for further details. If $GPn_FN = 0x002$, the respective pin output is CMOS.

The register fields that control the GPIO pins are described in Table 4-93.

Register Address	Bit	Label	Default	Description
R5888 (0x1700) GPIO1 CTRL 1	15	GPn_LVL	See Footnote 2	GPIO <i>n</i> level. Write to this bit to set a GPIO output. Read from this bit to read GPIO input level.
to R5918 (0x171E)				For output functions only, if GP <i>n</i> _POL is set, the GP <i>n</i> _LVL bit is the opposite logic level to the external pin.
GPIO16_CTRL1				Note that, if GPn_DIR = 0, the GPn_LVL bit is write-only.
	14	GPn_OP_CFG	0	GPIOn Output Configuration
				0 = CMOS
				1 = Open drain
				Note that, if $GPn_FN = 0x000$ or $0x002$, this bit has no effect on the GPIO <i>n</i> output. If GPn_FN = $0x000$, the pin configuration is set according to the applicable pin-specific function (see Table 4-95). If GPn_FN = $0x002$, the pin configuration is CMOS.
	13	GPn_DB	1	GPIOn Input Debounce
				0 = Disabled
				1 = Enabled
	12	GPn_POL	0	GPIOn Output Polarity Select
				0 = Noninverted (Active High)
				1 = Inverted (Active Low)
				Note that, if GP <i>n</i> _FN = 0x000 or 0x002, this bit has no effect on the GPIO <i>n</i> output.
	9:0	GP <i>n</i> _FN[9:0]	0x001	GPIOn Pin Function
				(see Table 4-94 for details)

Table 4-93. GPIO Control



Register Address	Bit	Label	Default	Description
R5889 (0x1701)	15	GPn_DIR	1	GPIOn Pin Direction
GPIO1_CTRL_2				0 = Output
to				1 = Input
R5919 (0x171F) GPIO16_CTRL2				Note that, if $GPn_FN = 0x000$ or $0x002$, this bit has no effect on the $GPIOn$ pin. If $GPn_FN = 0x000$, the pin direction is set according to the applicable pin-specific function (see Table 4-95). If $GPn_FN = 0x002$, the pin direction is set according to the DSP GPIO configuration.
	14	GPn_PU	1	GPIO <i>n</i> Pull-Up Enable
				0 = Disabled
				1 = Enabled
				Note: If GPn_PD and GPn_PU are both set, a bus keeper function is enabled on the respective GPIOn pin.
	13	GPn_PD	1	GPIO <i>n</i> Pull-Down Enable
				0 = Disabled
				1 = Enabled
				Note: If GP <i>n</i> _PD and GP <i>n</i> _PU are both set, a bus keeper function is enabled on the respective GPIO <i>n</i> pin.
R6848 (0x1AC0)	3:0	GP_DBTIME[3:0]	0x0	GPIO Input debounce time
GPIO_Debounce_				0x0 = 100 μs
Config				0x1 = 1.5 ms
				0x2 = 3 ms
				0x3 = 6 ms
				0x4 = 12 ms
				0x5 = 24 ms
				0x6 = 48 ms
				0x7 = 96 ms
				0x8 = 192 ms
				0x9 = 384 ms
				0xA = 768 ms
				0xB to 0xF = Reserved

Table 4-93. GPIO Control (Cont.)

1. n is a number (1–16) that identifies the individual GPIO.

The default value of GPn_LVL depends upon whether the pin is actively driven by another device. If the pin is actively driven, the bus keeper maintains this logic level. If the pin is not actively driven, the bus keeper may establish either a Logic 1 or Logic 0 as the initial input level.

4.14.2 GPIO Function Select

The available GPIO functions are described in Table 4-94. The function of each GPIO is set using GP n_FN , where n identifies the GPIO pin (1–16). Note that the respective GP n_DIR must also be set according to whether the function is an input or output.

GPn_FN	Valid On	Description	Comments
0x000	GPIO3–16 only	Pin-specific alternate function	Alternate functions supporting digital microphone, digital audio interface, master control interface, and PDM output functions.
0x001	All GPIOs (1–16)	Button-detect input/logic-level output	GP <i>n</i> _DIR = 0: GPIO pin logic level is set by GP <i>n</i> _LVL.
			GPn_DIR = 1: Button detect or logic level input.
0x002	All GPIOs (1–16)	DSP GPIO	Low latency input/output for DSP functions.
0x003	All GPIOs (1–16)	IRQ1 output	Interrupt (IRQ1) output
			0 = IRQ1 not asserted
			1 = IRQ1 asserted
0x004	All GPIOs (1–16)	IRQ2 output	Interrupt (IRQ2) output
			0 = IRQ2 not asserted
			1 = IRQ2 asserted
0x010	GPIO1–4 only	FLL1 clock	Clock output from FLL1
0x011	GPIO1–4 only	FLL2 clock	Clock output from FLL2

Table 4-94. GPIO Function Select



Table 4-94.	GPIO	Function	Select	(Cont.)
	01.10	i anouon	001000	

GPn_FN	Valid On	Description	Comments
0x018	GPIO1–4 only	FLL1 lock	Indicates FLL1 lock status
			0 = Not locked
			1 = Locked
0x019	GPIO1–4 only	FLL2 lock	Indicates FLL2 lock status
			0 = Not locked
			1 = Locked
0x040	GPIO1–4 only	OPCLK clock output	Configurable clock output derived from SYSCLK
0x041	GPIO1–4 only	OPCLK async clock output	Configurable clock output derived from ASYNCCLK
0x048	All GPIOs (1–16)	PWM1 output	Configurable PWM output PWM1
0x049	All GPIOs (1–16)	PWM2 output	Configurable PWM output PWM2
0x04C	All GPIOs (1–16)	S/PDIF output	IEC-60958-3-compatible S/PDIF output
0x088	GPIO1–4 only	ASRC1 IN1 lock	Indicates ASRC1 IN1 Lock status
			(ASRC IN1 paths convert from the ASRC1_RATE1 sample rate to the ASRC1 RATE2 sample rate.)
			0 = Not locked
			1 = Locked
0x089	GPIO1–4 only	ASRC1 IN2 lock	Indicates ASRC1 IN2 Lock status
			(ASRC IN2 paths convert from the ASRC1_RATE2 sample rate to the ASRC1_RATE1 sample rate.)
			0 = Not locked
			1 = Locked
0x140	All GPIOs (1–16)	Timer 1 status	Timer 1 Status
			A pulse is output after the respective timer reaches its final count value.
0x150	GPIO1–4 only	Event Log 1 FIFO not-empty status	Event Log 1 FIFO Not-Empty status
			0 = FIFO Empty
			1 = FIFO Not Empty
0x230	All GPIOs (1–16)	Alarm 1 Channel 1 status	Alarm 1 Channel 1 Status
			A pulse is output when the respective alarm-trigger conditions are met. The pulse duration is configurable.
0x231	All GPIOs (1–16)	Alarm 1 Channel 2 status	Alarm 1 Channel 2 Status
			A pulse is output when the respective alarm-trigger conditions are met. The pulse duration is configurable.
0x232	All GPIOs (1–16)	Alarm 1 Channel 3 status	Alarm 1 Channel 3 Status
			A pulse is output when the respective alarm-trigger conditions are met. The pulse duration is configurable.
0x233	All GPIOs (1–16)	Alarm 1 Channel 4 status	Alarm 1 Channel 4 Status
			A pulse is output when the respective alarm-trigger conditions are met. The pulse duration is configurable.
0x280	GPIO3 or GPIO10	Auxiliary PDM clock input/output	Auxiliary PDM interface clock
0x281	GPIO4 or GPIO9	Auxiliary PDM data output	Auxiliary PDM interface data

4.14.3 Pin-Specific Alternate Function—GP*n*_FN = 0x000

The CS42L92 provides two dedicated GPIO pins (1–2). The remaining 14 GPIOs are multiplexed with the pin-specific functions listed in Table 4-95. The alternate functions are selected by setting the respective GPn_FN fields to 0x000, as described in Section 4.14.1. Note that each function is unique to the associated pin and can be supported only on that pin.

If the alternate function is selected on a GPIO pin, the pin direction (input or output) and the output driver configuration (CMOS or open drain) are set automatically as described in Table 4-95. The respective GP*n*_DIR and GP*n*_OP_CFG bits have no effect in this case.

Name	Condition	Description	Direction	Output Driver Configuration
AIF1BCLK/GPIO6	GP6_FN = 0x000	Audio Interface 1 bit clock	Digital I/O	CMOS
AIF1LRCLK/GPIO8	GP8_FN = 0x000	Audio Interface 1 left/right clock	Digital I/O	CMOS
AIF1RXDAT/GPI07	GP7_FN = 0x000	Audio Interface 1 RX digital audio data	Digital input	—

	Table 4-95.	GPIO Alternate	Functions
--	-------------	-----------------------	-----------



Name	Condition	Description	Direction	Output Driver Configuration
AIF1TXDAT/GPIO5	GP5_FN = 0x000	Audio Interface 1 TX digital audio data	Digital output	CMOS
AIF2BCLK/GPIO10	GP10_FN = 0x000	Audio Interface 2 bit clock	Digital I/O	CMOS
AIF2LRCLK/GPI012	GP12_FN = 0x000	Audio Interface 2 left/right clock	Digital I/O	CMOS
AIF2RXDAT/GPIO11	GP11_FN = 0x000	Audio Interface 2 RX digital audio data	Digital input	—
AIF2TXDAT/GPIO9	GP9_FN = 0x000	Audio Interface 2 TX digital audio data	Digital output	CMOS
AIF3BCLK/GPI014	GP14_FN = 0x000	Audio Interface 3 bit clock	Digital I/O	CMOS
AIF3LRCLK/GPIO16	GP16_FN = 0x000	Audio Interface 3 left/right clock	Digital I/O	CMOS
AIF3RXDAT/GPIO15	GP15_FN = 0x000	Audio Interface 3 RX digital audio data	Digital input	—
AIF3TXDAT/GPI013	GP13_FN = 0x000	Audio Interface 3 TX digital audio data	Digital output	CMOS
SPKCLK/GPIO3	GP3_FN = 0x000	Digital speaker (PDM) clock	Digital output	CMOS
SPKDAT/GPIO4	GP4_FN = 0x000	Digital speaker (PDM) data	Digital output	CMOS

Table 4-95. GPIO Alternate Functions (Cont.)

Note that if the JTAG interface is enabled, the GPIO13–15 pins are configured as a JTAG interface. Under these conditions, the respective GPIO configuration registers have no effect, and the GPIO pins cannot be assigned any other function. See Section 4.20 for details of the JTAG interface.

4.14.4 Button Detect (GPIO Input)—GPn_FN = 0x001

Button-detect functionality can be selected on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1. The same functionality can be used to support a jack-detect input function.

It is recommended to enable the GPIO input debounce feature when using GPIOs as button input or jack-detect input.

The GP n_LVL fields may be read to determine the logic levels on a GPIO input, after the selectable debounce controls. Note that GP n_LVL is not affected by the GP n_POL bit.

The debounced GPIO signals are also inputs to the interrupt-control circuit. An interrupt event is triggered on the rising and falling edges of the GPIO input. The associated interrupt bits are latched once set; they can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.

4.14.5 Logic 1 and Logic 0 Output (GPIO Output)—GPn_FN = 0x001

The CS42L92 can be programmed to drive a logic high or logic low level on a GPIO pin by selecting the GPIO Output function as described in Section 4.14.1.

The output logic level is selected using the respective GPn_LVL bit. Note that, if a GPIO pin is configured as an output, the respective GPn_LVL bits are write-only.

The polarity of the GPIO output can be inverted using the GP n_POL bits. If GP n_POL = 1, the external output is the opposite logic level to GP n_LVL .

4.14.6 DSP GPIO (Low-Latency DSP Input/Output)—GPn_FN = 0x002

The DSP GPIO function provides an advanced I/O capability for signal-processing applications. The DSP GPIO pins are accessed using maskable sets of I/O control registers; this allows the selected combinations of GPIOs to be controlled with ease, regardless of how the allocation of GPIO pins has been implemented in hardware.

The DSP GPIO function is selected by setting the respective GPIO fields as described in Section 4.14.1.

A full description of the DSP GPIO function is provided in Section 4.5.4.

Note that, if GPn_FN is set to 0x002, the respective pin direction (input or output) is set according to the DSP GPIO configuration for that pin—the GPn_DIR control bit has no effect in this case.



4.14.7 Interrupt (IRQ) Status Output—GPn_FN = 0x003, 0x004

The CS42L92 has an interrupt controller, which can be used to indicate when any selected interrupt events occur. Individual interrupts may be masked in order to configure the interrupt as required. See Section 4.15 for a full definition of all supported interrupt events.

The interrupt controller supports two separate interrupt request (IRQ) outputs. The IRQ1 or IRQ2 status may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.

Note that the IRQ1 status is output on the \overline{IRQ} pin at all times.

4.14.8 Frequency-Locked Loop (FLL) Clock Output—GPn_FN = 0x010, 0x011

Clock outputs derived from the FLLs may be output on a GPIO pin. The GPIO output from each FLL*n* (FLL1 or FLL2) is controlled by the respective FLL*n*_GPCLK_DIV and FLL*n*_GPCLK_ENA fields, as described in Table 4-96.

It is recommended to disable the clock output (FLL*n*_GPCLK_ENA = 0) before making any change to the respective FLL*n*_GPCLK_DIV field.

Note that FLL*n*_GPCLK_DIV and FLL*n*_GPCLK_ENA affect the GPIO outputs only; they do not affect the FLL frequency. The maximum output frequency supported for GPIO output is noted in Table 3-10.

The FLL clock outputs may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.

See Section 4.16 for details of the CS42L92 system clocking and how to configure the FLLs.

Register Address	Bit	Label	Default	Description
R398 (0x018E)	7:1	FLL1_GPCLK_	0x02	FLL1 GPIO Clock Divider
FLL1_GPIO_Clock		DIV[6:0]		0x00 = Reserved
				0x01 = Reserved
				0x02 = Divide by 2
				0x03 = Divide by 3
				0x04 = Divide by 4
				0x7F = Divide by 127
				(F _{GPIO} = F _{FLL} /FLL1_GPCLK_DIV)
	0	FLL1_GPCLK_	0	FLL1 GPIO Clock Enable
		ENA		0 = Disabled
				1 = Enabled
R430 (0x01AE)	7:1	FLL2_GPCLK_	0x02	FLL2 GPIO Clock Divider
FLL2_GPIO_Clock		DIV[6:0]		0x00 = Reserved
				0x01 = Reserved
				0x02 = Divide by 2
				0x03 = Divide by 3
				0x04 = Divide by 4
				0x7F = Divide by 127
				(F _{GPIO} = F _{FLL} /FLL2_GPCLK_DIV)
	0	FLL2_GPCLK_	0	FLL2 GPIO Clock Enable
		ENA		0 = Disabled
				1 = Enabled

 Table 4-96.
 FLL Clock Output Control

4.14.9 Frequency-Locked Loop (FLL) Status Output—GPn_FN = 0x018, 0x019

The CS42L92 provides FLL status flags, which may be used to control other events. The FLL lock signals indicate whether FLL lock has been achieved. See Section 4.16.8 for details of the FLLs.

The FLL lock signals may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.



The FLL lock signals are inputs to the interrupt controller circuit. An interrupt event is triggered on the rising and falling edges of these signals. The associated interrupt bits are latched once set; they can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.

4.14.10 OPCLK and OPCLK_ASYNC Clock Output—GPn_FN = 0x040, 0x041

A clock output (OPCLK) derived from SYSCLK can be output on a GPIO pin. The OPCLK frequency is controlled by OPCLK_DIV and OPCLK_SEL. The OPCLK output is enabled by setting OPCLK_ENA, as described in Table 4-97.

A clock output (OPCLK_ASYNC) derived from ASYNCCLK can be output on a GPIO pin. The OPCLK_ASYNC frequency is controlled by OPCLK_ASYNC_DIV and OPCLK_ASYNC_SEL. The OPCLK_ASYNC output is enabled by setting OPCLK_ASYNC_ENA.

It is recommended to disable the clock output (OPCLK_ENA = 0 or OPCLK_ASYNC_ENA = 0) before making any change to the respective OPCLK_DIV, OPCLK_SEL, OPCLK_ASYNC_DIV, or OPCLK_ASYNC_SEL fields.

The OPCLK or OPCLK_ASYNC clock can be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.

Note that the OPCLK source frequency cannot be higher than the SYSCLK frequency. The OPCLK_ASYNC source frequency cannot be higher than the ASYNCCLK frequency. The maximum output frequency supported for GPIO output is noted in Table 3-10.

See Section 4.16 for details of the system clocks (SYSCLK and ASYNCCLK).

Register Address	Bit	Label	Default	Description
R329 (0x0149)	15	OPCLK_ENA	0	OPCLK Enable
Output_system_				0 = Disabled
clock				1 = Enabled
	7:3	OPCLK_DIV[4:0]	0x00	OPCLK Divider
				0x02 = Divide by 2
				0x04 = Divide by 4
				0x06 = Divide by 6
				(even numbers only)
				0x1E = Divide by 30
				Note that only even numbered divisions (2, 4, 6, etc.) are valid selections.
				All other codes are reserved when the OPCLK signal is enabled.
	2:0	OPCLK_SEL[2:0]	000	OPCLK Source Frequency
				000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related SYSCLK rates only (i.e., SAMPLE_RATE_ <i>n</i> = 01XXX).
				The OPCLK Source Frequency must be less than or equal to the SYSCLK frequency.

Table 4-97. OPCLK Control



Register Address	Bit	Label	Default	Description		
R330 (0x014A)	15	OPCLK_ASYNC_	0	OPCLK_ASYNC Enable		
Output_async_		ENA		0 = Disabled		
lock				1 = Enabled		
	7:3	OPCLK_ASYNC_	0x00	OPCLK_ASYNC Divider		
		DIV[4:0]		0x02 = Divide by 2		
				0x04 = Divide by 4		
				0x06 = Divide by 6		
				(even numbers only)		
				0x1E = Divide by 30		
				Note that only even numbered divisions (2, 4, 6, etc.) are valid selections.		
				All other codes are reserved when the OPCLK_ASYNC signal is enabled.		
	2:0	OPCLK_ASYNC_	000	OPCLK_ASYNC Source Frequency		
		SEL[2:0]		000 = 6.144 MHz (5.6448 MHz)		
				001 = 12.288 MHz (11.2896 MHz)		
				010 = 24.576 MHz (22.5792 MHz)		
				011 = 49.152 MHz (45.1584 MHz)		
				All other codes are reserved		
				The frequencies in brackets apply for 44.1 kHz–related ASYNCCLK rates only (i.e., ASYNC_SAMPLE_RATE_n = 01XXX).		
				The OPCLK_ASYNC Source Frequency must be less than or equal to the ASYNCCLK frequency.		

Table 4-97. OPCLK Control (Cont.)

4.14.11 Pulse-Width Modulation (PWM) Signal Output—GPn_FN = 0x048, 0x049

The CS42L92 incorporates two PWM signal generators, which can be enabled as GPIO outputs. The duty cycle of each PWM signal can be modulated by an audio source, or can be set to a fixed value using a control register setting.

The PWM outputs may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.

See Section 4.3.12 for details of how to configure the PWM signal generators.

4.14.12 S/PDIF Audio Output—GPn_FN = 0x04C

The CS42L92 incorporates an IEC-60958-3–compatible S/PDIF transmitter, which can be selected as a GPIO output. The S/PDIF transmitter supports stereo audio channels and allows full control over the S/PDIF validity bits and channel status information.

The S/PDIF signal may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.

See Section 4.3.8 for details of how to configure the S/PDIF output generator.

4.14.13 ASRC Lock Status Output—GPn_FN = 0x088, 0x089

The CS42L92 provides ASRC status flags, which may be used to control other events. The ASRC-lock signals indicate whether ASRC lock has been achieved. See Section 4.3.15 for details of the ASRCs.

The ASRC lock signals may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1.

The ASRC lock signals are inputs to the interrupt control circuit. An interrupt event is triggered on the rising and falling edges of the ASRC lock signals. The associated interrupt bits are latched once set; they can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.



4.14.14 General-Purpose Timer Status Output—GPn_FN = 0x140

The general-purpose timer can count up or down, and supports continuous or single count modes. A status output, indicating the progress of the timer, is provided. See Section 4.5.3 for details of the general-purpose timer.

A logic signal from the general-purpose timer may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1. This logic signal is pulsed high whenever the timer reaches its final count value.

The general-purpose timer also provides input to the interrupt control circuit. An interrupt event is triggered whenever the timer reaches its final count value. The associated interrupt bit is latched once set; it can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.

4.14.15 Event Logger FIFO Buffer Status Output—GPn_FN = 0x150

The event logger incorporates a 16-stage FIFO buffer, in which any detected events (signal transitions) are recorded. A status output for the FIFO buffer is provided. See Section 4.5.1 for details of the event logger.

A logic signal from the event logger may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1. This logic signal is set high whenever the FIFO not-empty condition is true.

The event logger also provides input to the interrupt control circuit. An interrupt event is triggered whenever the FIFO condition occurs. The associated interrupt bit is latched once set; it can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.

4.14.16 Alarm Generator Status Output—GPn_FN = 0x230, 0x231, 0x232, 0x233

The CS42L92 incorporates four alarm-generator circuits that are associated with the general-purpose timer. A status output is provided by each alarm; these can be used to indicate one-off events, or can be configured for cyclic (repeated) triggers. See Section 4.5.2 for details of the alarm-control circuits.

The alarm status may be output directly on a GPIO pin by setting the respective GPIO fields as described in Section 4.14.1. The alarm status is asserted when the respective alarm-trigger conditions are met. The signal is asserted for a duration that is configurable as described in Section 4.5.2.1.

The alarm generators also provide input to the interrupt control circuit. An interrupt event is triggered whenever the alarm-trigger conditions are met. The associated interrupt bit is latched once set; it can be polled at any time or used to control the IRQ signal. See Section 4.15 for details of the interrupt event handling.

4.14.17 Auxiliary PDM Interface—GPn_FN = 0x280, 0x281

The CS42L92 provides an auxiliary PDM interface that can be used to provide an audio path between an analog microphone connected to the CS42L92 and a digital input to an external audio processor. The external connections to the auxiliary PDM interface are supported on GPIO pins as follows:

- In Master Mode (AUXPDM1_MSTR = 1), the CLK output can be configured on the GPIO3 or GPIO10 pins. The DAT
 output can be configured on GPIO4 or GPIO9.
- In Slave Mode (AUXPDM1_MSTR = 0), the CLK input is supported on GPIO10 only. The DAT output is supported on GPIO9 only.

The applicable GPIO pins are configured by setting the respective GPIO fields as described in Section 4.14.1.

See Section 4.2.10 for details of how to configure the auxiliary PDM interface.

4.14.18 General-Purpose Switch

The CS42L92 provides two general-purpose switches, which can be used as controllable analog switches for external functions. The switches support bidirectional analog operation, offering flexibility in the potential circuit applications. Refer to Table 3-2 and Table 3-10 for further details. Note that this feature is entirely independent of the GPIO*n* pins.

- The GP1 switch is implemented between the GPSW1P and GPSW1N pins; it is configured using SW1_MODE.
- The GP2 switch is implemented between the GPSW2P and GPSW2N pins; it is configured using SW2_MODE.



The SW*n*_MODE fields allow the switches to be disabled, enabled, or synchronized to the MICDET clamp status, as described in Table 4-98.

The switches can be used in conjunction with the MICDET clamp function to suppress pops and clicks associated with jack insertion and removal. An example circuit is shown in Fig. 4-60 within Section 4.12.2. Note that the MICDET clamp function must also be configured appropriately when using this method of pop suppression.

Register Address	Bit	Label	Default	Description
R712 (0x02C8)	3:2	SW2_MODE[1:0]	00	General-purpose Switch 2 control
GP_Switch_1				00 = Disabled (switch open)
				01 = Enabled (switch closed)
				10 = Enabled if MICDET clamp status is set
				11 = Enabled if MICDET clamp status is clear
	1:0	SW1_MODE[1:0]	00	General-purpose Switch 1 control
				00 = Disabled (switch open)
				01 = Enabled (switch closed)
				10 = Enabled if MICDET clamp status is set
				11 = Enabled if MICDET clamp status is clear

Table 4-98.	General-Purpose	Switch C	ontrol
		• •	•••••

4.15 Interrupts

The interrupt controller has multiple inputs. These include the jack detect and GPIO input pins, DSP_IRQn flags, headphone/accessory detection, FLL/ASRC lock detection, and status flags from DSP peripheral functions. See Table 4-99 and Table 4-100 for a full definition of the interrupt controller inputs. Any combination of these inputs can be used to trigger an interrupt request event.

The interrupt controller supports two sets of interrupt registers. This allows two separate interrupt request (IRQ) outputs to be generated, and for each IRQ to report a different set of input or status conditions.

For each interrupt request (IRQ1 and IRQ2) output, there is an interrupt register field associated with each interrupt input. These fields are asserted whenever a logic edge is detected on the respective input. Some inputs are triggered on rising edges only; some are triggered on both edges. Separate rising and falling interrupt bits are provided for the JD1 and JD2 signals. The interrupt register fields for IRQ1 are described in Table 4-99. The interrupt register fields for IRQ2 are described in Table 4-100. The interrupt flags can be polled at any time or in response to the interrupt request output being signaled via the IRQ pin or a GPIO pin.

All interrupts are edge triggered, as noted above. Many are triggered on both the rising and falling edges and, therefore, the interrupt bits cannot indicate which edge has been detected. The raw status fields described in Table 4-99 and Table 4-100 indicate the current value of the corresponding inputs to the interrupt controller. Note that the raw status bits associated with IRQ1 and IRQ2 provide the same information. The status of any GPIO (or DSP GPIO) inputs can also be read using the GPIO (or DSP GPIO) control fields, as described in Table 4-93 and Table 4-41.

Individual mask bits can enable or disable different functions from the interrupt controller. The mask bits are described in Table 4-99 (for IRQ1) and Table 4-100 (for IRQ2). Note that a masked interrupt input does not assert the corresponding interrupt register field and does not cause the associated interrupt request output to be asserted.

The interrupt request outputs represent the logical OR of the associated interrupt registers. IRQ1 is derived from the x_EINT1 registers; IRQ2 is derived from the x_EINT2 registers. The interrupt register fields are latching fields and, once they are set, they are not reset until a 1 is written to the respective bits. The interrupt request outputs are not reset until each of the associated interrupts has been reset.

A debounce circuit can be enabled on any GPIO input, to avoid false event triggers. This is enabled on each pin using the fields described in Table 4-93. The GPIO debounce circuit uses the 32-kHz clock, which must be enabled whenever the GPIO debounce function is required.

A debounce circuit is always enabled on the FLL status inputs—either the 32-kHz clock or the SYSCLK signal must be enabled to trigger an interrupt from the FLL status inputs. Note that the raw status fields (described in Table 4-99 and Table 4-100) are valid without clocking; these fields can be used to provide FLL status readback if system clocks are not available.



The IRQ outputs can be globally masked using the IM_IRQ1 and IM_IRQ2 bits. When not masked, the IRQ status can be read from IRQ1_STS and IRQ2_STS for the respective IRQ outputs.

The IRQ1 output is provided externally on the IRQ pin. Under default conditions, this output is active low. The polarity can be inverted using IRQ_POL. The IRQ output can be either CMOS driven or open drain; this is selected using the IRQ_OP_CFG bit. The IRQ output is active low and is referenced to the DBVDD power domain.

The IRQ2 status can be used to trigger DSP firmware execution; see Section 4.4. This allows the DSP firmware execution to be linked to external events (e.g., jack detection, or GPIO input), or to any of the status conditions flagged by the interrupt registers.

The IRQ1 and IRQ2 signals may be output on a GPIO pin; see Section 4.14.

The CS42L92 interrupt controller circuit is shown in Fig. 4-64. (Note that not all interrupt inputs are shown.) The control fields associated with IRQ1 and IRQ2 are described in Table 4-99 and Table 4-100 respectively. The global interrupt mask bits, status bits, and output configuration fields are described Table 4-101.

Note that, under default register conditions, the boot done status is the only unmasked interrupt source; a falling edge on the IRQ pin indicates completion of the boot sequence.

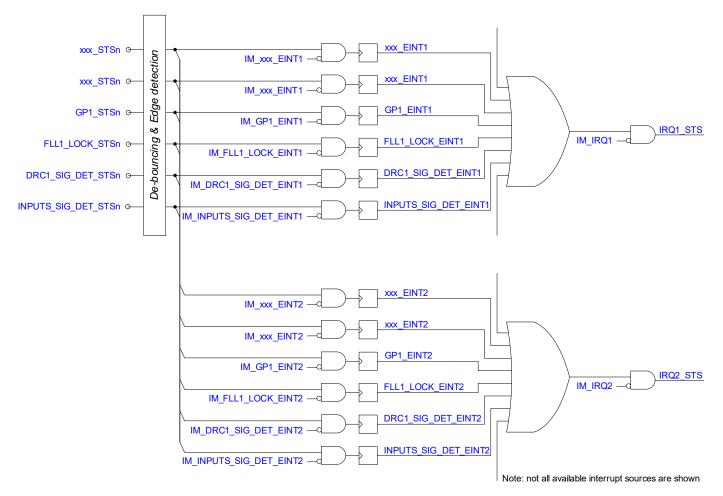


Figure 4-64. Interrupt Controller



The IRQ1 interrupt, mask, and status control registers are described in Table 4-99.

Table 4-99.	Interrupt 1 Contro	I Registers
-------------	--------------------	-------------

Register Address	Bit	Label	Default	Description
R6144 (0x1800)	12	CTRLIF_ERR_EINT1	0	Control Interface Error Interrupt (Rising edge triggered)
IRQ1_Status_1				Note: Cleared when a 1 is written.
	9	SYSCLK_FAIL_EINT1	0	SYSCLK Fail Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	7	BOOT_DONE_EINT1	0	Boot Done Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6145 (0x1801)	14	DSPCLK_ERR_EINT1	0	DSPCLK Error Interrupt (Rising edge triggered)
IRQ1_Status_2				Note: Cleared when a 1 is written.
	13	ASYNCCLK_ERR_EINT1	0	ASYNCCLK Error Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	12	SYSCLK_ERR_EINT1	0	SYSCLK Error Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	9	FLL2_LOCK_EINT1	0	FLL2 Lock Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	8	FLL1_LOCK_EINT1	0	FLL1 Lock Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	6	FLL2_REF_LOST_EINT1	0	FLL2 Reference Lost Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	5	FLL1_REF_LOST_EINT1	0	FLL1 Reference Lost Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6149 (0x1805)	9	MICDET2_EINT1	0	Mic/Accessory Detect 2 Interrupt (Detection event triggered)
IRQ1_Status_6				Note: Cleared when a 1 is written.
	8	MICDET1_EINT1	0	Mic/Accessory Detect 1 Interrupt (Detection event triggered)
				Note: Cleared when a 1 is written.
	0	HPDET_EINT1	0	Headphone Detect Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6150 (0x1806)	11	MICD_CLAMP2_FALL_EINT1	0	MICDET Clamp 2 Interrupt (Falling edge triggered)
IRQ1_Status_7				Note: Cleared when a 1 is written.
	10	MICD_CLAMP2_RISE_EINT1	0	MICDET Clamp 2 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	9	JD3_FALL_EINT1	0	JD3 Interrupt (Falling edge triggered)
				Note: Cleared when a 1 is written.
	8	JD3_RISE_EINT1	0	JD3 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	5	MICD_CLAMP1_FALL_EINT1	0	MICDET Clamp 1 Interrupt (Falling edge triggered)
				Note: Cleared when a 1 is written.
	4	MICD_CLAMP1_RISE_EINT1	0	MICDET Clamp 1 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	JD2_FALL_EINT1	0	JD2 Interrupt (Falling edge triggered)
				Note: Cleared when a 1 is written.
	2	JD2_RISE_EINT1	0	JD2 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	JD1_FALL_EINT1	0	JD1 Interrupt (Falling edge triggered)
				Note: Cleared when a 1 is written.
	0	JD1_RISE_EINT1	0	JD1 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6152 (0x1808)	9	ASRC1_IN2_LOCK_EINT1	0	ASRC1 IN2 Lock Interrupt (Rising and falling edge triggered)
IRQ1_Status_9				Note: Cleared when a 1 is written.
	8	ASRC1_IN1_LOCK_EINT1	0	ASRC1 IN1 Lock Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	2	INPUTS_SIG_DET_EINT1	0	Input Path Signal-Detect Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	1	DRC2_SIG_DET_EINT1	0	DRC2 Signal-Detect Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	0	DRC1_SIG_DET_EINT1	0	DRC1 Signal-Detect Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
R6154 (0x180A)	15	DSP_IRQ16_EINT1	0	DSP IRQ16 Interrupt (Rising edge triggered)
IRQ1_Status_11				Note: Cleared when a 1 is written.
	14	DSP_IRQ15_EINT1	0	DSP IRQ15 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	13	DSP_IRQ14_EINT1	0	DSP IRQ14 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	12	DSP_IRQ13_EINT1	0	DSP IRQ13 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	11	DSP_IRQ12_EINT1	0	DSP IRQ12 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	10	DSP_IRQ11_EINT1	0	DSP IRQ11 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	9	DSP IRQ10 EINT1	0	DSP IRQ10 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	8	DSP IRQ9 EINT1	0	DSP IRQ9 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	7	DSP_IRQ8_EINT1	0	DSP IRQ8 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	6	DSP_IRQ7_EINT1	0	DSP IRQ7 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	5	DSP IRQ6 EINT1	0	DSP IRQ6 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	4	DSP IRQ5 EINT1	0	DSP IRQ5 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	DSP IRQ4 EINT1	0	DSP IRQ4 Interrupt (Rising edge triggered)
			-	Note: Cleared when a 1 is written.
	2	DSP IRQ3 EINT1	0	DSP IRQ3 Interrupt (Rising edge triggered)
	-		-	Note: Cleared when a 1 is written.
	1	DSP_IRQ2_EINT1	0	DSP IRQ2 Interrupt (Rising edge triggered)
	•			Note: Cleared when a 1 is written.
	0	DSP_IRQ1_EINT1	0	DSP IRQ1 Interrupt (Rising edge triggered)
	0		Ŭ	Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6155 (0x180B)	9	HP4R_SC_EINT1	0	HPOUT4R Short Circuit Interrupt (Rising edge triggered)
IRQ1_Status_12				Note: Cleared when a 1 is written.
	8	HP4L_SC_EINT1	0	HPOUT4L Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	5	HP3R_SC_EINT1	0	HPOUT3R Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	4	HP3L_SC_EINT1	0	HPOUT3L Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	HP2R_SC_EINT1	0	HPOUT2R Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	2	HP2L_SC_EINT1	0	HPOUT2L Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	HP1R_SC_EINT1	0	HPOUT1R Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	HP1L_SC_EINT1	0	HPOUT1L Short Circuit Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6156 (0x180C)	5	HP3R_ENABLE_DONE_EINT1	0	HPOUT3R/HPOUT4R Enable Interrupt (Rising edge triggered)
IRQ1_Status_13				Note: Cleared when a 1 is written.
	4	HP3L_ENABLE_DONE_EINT1	0	HPOUT3L/HPOUT4L Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	HP2R_ENABLE_DONE_EINT1	0	HPOUT2R Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	2	HP2L ENABLE DONE EINT1	0	HPOUT2L Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	HP1R ENABLE DONE EINT1	0	HPOUT1R Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	HP1L_ENABLE_DONE_EINT1	0	HPOUT1L Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6157 (0x180D)	5	HP3R_DISABLE_DONE_EINT1	0	HPOUT3R/HPOUT4R Disable Interrupt (Rising edge triggered)
IRQ1_Status_14				Note: Cleared when a 1 is written.
	4	HP3L_DISABLE_DONE_EINT1	0	HPOUT3L/HPOUT4L Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	HP2R_DISABLE_DONE_EINT1	0	HPOUT2R Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	2	HP2L_DISABLE_DONE_EINT1	0	HPOUT2L Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	HP1R DISABLE DONE EINT1	0	HPOUT1R Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	HP1L DISABLE DONE EINT1	0	HPOUT1L Disable Interrupt (Rising edge triggered)
			-	Note: Cleared when a 1 is written.
R6158 (0x180E)	12	DFC SATURATE EINT1	0	DFC Saturate Interrupt (Rising edge triggered)
IRQ1_Status_15				Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6160 (0x1810)	15	GP16_EINT1	0	GPIO16 Interrupt (Rising and falling edge triggered)
IRQ1_Status_17				Note: Cleared when a 1 is written.
	14	GP15_EINT1	0	GPIO15 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	13	GP14_EINT1	0	GPIO14 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	12	GP13_EINT1	0	GPIO13 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	11	GP12_EINT1	0	GPIO12 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	10	GP11_EINT1	0	GPIO11 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	9	GP10_EINT1	0	GPIO10 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	8	GP9_EINT1	0	GPIO9 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	7	GP8 EINT1	0	GPIO8 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	6	GP7_EINT1	0	GPIO7 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	5	GP6_EINT1	0	GPIO6 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	4	GP5 EINT1	0	GPIO5 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	3	GP4 EINT1	0	GPIO4 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	2	GP3 EINT1	0	GPIO3 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	1	GP2 EINT1	0	GPIO2 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	0	GP1_EINT1	0	GPIO1 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
R6164 (0x1814)	0	TIMER1_EINT1	0	Timer 1 Interrupt (Rising edge triggered)
IRQ1 Status 21		_		Note: Cleared when a 1 is written.
R6165 (0x1815)	0	EVENT1_NOT_EMPTY_EINT1	0	Event Log 1 FIFO Not Empty Interrupt (Rising edge triggered)
IRQ1_Status_22				Note: Cleared when a 1 is written.
R6166 (0x1816)	0	EVENT1 FULL EINT1	0	Event Log 1 FIFO Full Interrupt (Rising edge triggered)
IRQ1 Status 23				Note: Cleared when a 1 is written.
R6167 (0x1817)	0	EVENT1 WMARK EINT1	0	Event Log 1 FIFO Watermark Interrupt (Rising edge triggered)
IRQ1_Status_24	-			Note: Cleared when a 1 is written.
R6168 (0x1818)	0	DSP1 DMA EINT1	0	DSP1 DMA Interrupt (Rising edge triggered)
IRQ1_Status_25	Ũ		Ũ	Note: Cleared when a 1 is written.
R6170 (0x181A)	0	DSP1 START1 EINT1	0	DSP1 Start 1 Interrupt (Rising edge triggered)
	0		Ŭ	Note: Cleared when a 1 is written.
IRQ1_Status_27	0	DSP1 START2 EINT1		DSP1 Start 2 Interrupt (Rising edge triggered)
R6171 (0x181B)	U	DOF I_DIARIZ_EINTI	0	Note: Cleared when a 1 is written.
IRQ1_Status_28				
R6173 (0x181D)	0	DSP1_BUSY_EINT1	0	DSP1 Busy Interrupt (Rising edge triggered)
IRQ1_Status_30				Note: Cleared when a 1 is written.
R6176 (0x1820)	0	DSP1_BUS_ERR_EINT1	0	DSP1 Bus Error Interrupt (Rising edge triggered)
IRQ1_Status_33				Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6179 (0x1823)	3	TIMER_ALM1_CH4_EINT1	0	Alarm 1 Channel 4 Interrupt (Rising edge triggered)
IRQ1_Status_36				Note: Cleared when a 1 is written.
	2	TIMER_ALM1_CH3_EINT1	0	Alarm 1 Channel 3 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	TIMER_ALM1_CH2_EINT1	0	Alarm 1 Channel 2 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	TIMER_ALM1_CH1_EINT1	0	Alarm 1 Channel 1 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6208 (0x1840)		IM_*	See	For each x_EINT1 interrupt bit in R6144 to R6179, a
to			Footnote ¹	corresponding mask bit (IM_*) is provided in R6208 to R6243.
R6243 (0x1863)				The mask bits are coded as follows:
				0 = Do not mask interrupt
				1 = Mask interrupt
R6272 (0x1880)	12	CTRLIF_ERR_STS1	0	Control Interface Error Status
IRQ1_Raw_				0 = Normal
Status_1				1 = Control Interface Error
	7	BOOT_DONE_STS1	0	Boot Status
				0 = Busy (boot sequence in progress)
				1 = Idle (boot sequence completed)
				Control register writes should not be attempted until Boot Sequence has completed.
R6273 (0x1881)	14	DSPCLK_ERR_STS1	0	DSPCLK Error Interrupt Status
IRQ1_Raw_				0 = Normal
Status_2				1 = Insufficient DSPCLK cycles for one or more of the requested DSP1 clock frequencies
-	13	ASYNCCLK_ERR_STS1	0	ASYNCCLK Error Interrupt Status
				0 = Normal
				1 = Insufficient ASYNCCLK cycles for the requested signal path functionality
	12	SYSCLK_ERR_STS1	0	SYSCLK Error Interrupt Status
				0 = Normal
				1 = Insufficient SYSCLK cycles for the requested signal path functionality
	9	FLL2_LOCK_STS1	0	FLL2 Lock Status
				0 = Not locked
				1 = Locked
	8	FLL1_LOCK_STS1	0	FLL1 Lock Status
				0 = Not locked
				1 = Locked
	6	FLL2_REF_LOST_STS1	0	FLL2 Reference Lost Status
				0 = Normal
				1 = Reference Lost
	5	FLL1_REF_LOST_STS1	0	FLL1 Reference Lost Status
				0 = Normal
				1 = Reference Lost



Register Address	Bit	Label	Default	Description
R6278 (0x1886)	8	JD3_STS1	0	JACKDET3 input status
IRQ1_Raw_				0 = Jack not detected
Status_7				1 = Jack is detected
				(Assumes the JACKDET3 pin is pulled low on jack insertion.)
	4	MICD_CLAMP_STS1	0	MICDET Clamp status
				0 = Clamp disabled
				1 = Clamp enabled
				Separate _STS bits are not provided for each clamp—it is assumed that a maximum of one clamp is active at any time. The clamp override condition (MICD_CLAMP <i>n</i> _OVD = 1) is not indicated.
	2	JD2_STS1	0	JACKDET2 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET2 pin is pulled low on jack insertion.)
	0	JD1_STS1	0	JACKDET1 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET1 pin is pulled low on jack insertion.)
R6280 (0x1888)	9	ASRC1_IN2_LOCK_STS1	0	ASRC1 IN2 Lock Status
IRQ1_Raw_				0 = Not locked
Status_9				1 = Locked
	8	ASRC1_IN1_LOCK_STS1	0	ASRC1 IN1 Lock Status
				0 = Not locked
				1 = Locked
	2	INPUTS_SIG_DET_STS1	0	Input Path Signal-Detect Status
				0 = Normal
				1 = Signal detected
	1	DRC2_SIG_DET_STS1	0	DRC2 Signal-Detect Status
				0 = Normal
				1 = Signal detected
	0	DRC1_SIG_DET_STS1	0	DRC1 Signal-Detect Status
				0 = Normal
				1 = Signal detected



Register Address	Bit	Label	Default	Description
R6283 (0x188B)	9	HP4R_SC_STS1	0	HPOUT4R Short Circuit Status
IRQ1_Raw_				0 = Normal
Status_12				1 = Short Circuit detected
	8	HP4L_SC_STS1	0	HPOUT4L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	5	HP3R_SC_STS1	0	HPOUT3R Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	4	HP3L_SC_STS1	0	HPOUT3L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	3	HP2R_SC_STS1	0	HPOUT2R Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	2	HP2L SC STS1	0	HPOUT2L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	1	HP1R_SC_STS1	0	HPOUT1R Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	0	HP1L SC STS1	0	HPOUT1L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
R6284 (0x188C)	5	HP3R ENABLE DONE STS1	0	HPOUT3R/HPOUT4R Enable Status
IRQ1 Raw			_	0 = Busy (sequence in progress)
Status_13				1 = Idle (sequence completed)
	4	HP3L ENABLE DONE STS1	0	HPOUT3L/HPOUT4L Enable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	3	HP2R ENABLE DONE STS1	0	HPOUT2R Enable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	2	HP2L ENABLE DONE STS1	0	HPOUT2L Enable Status
			_	0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	1	HP1R ENABLE DONE STS1	0	HPOUT1R Enable Status
	-		-	0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	0	HP1L ENABLE DONE STS1	0	HPOUT1L Enable Status
	•			0 = Busy (sequence in progress)
				1 = Idle (sequence completed)



Register Address	Bit	Label	Default	Description
R6285 (0x188D)	5	HP3R_DISABLE_DONE_STS1	0	HPOUT3R/HPOUT4R Disable Status
IRQ1_Raw_				0 = Busy (sequence in progress)
Status_14				1 = Idle (sequence completed)
	4	HP3L_DISABLE_DONE_STS1	0	HPOUT3L/HPOUT4L Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	3	HP2R_DISABLE_DONE_STS1	0	HPOUT2R Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	2	HP2L_DISABLE_DONE_STS1	0	HPOUT2L Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	1	HP1R_DISABLE_DONE_STS1	0	HPOUT1R Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
·	0	HP1L_DISABLE_DONE_STS1	0	HPOUT1L Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
R6288 (0x1890)	15	GP16 STS1	0	GPIOn Input status. Reads back the logic level of GPIOn.
IRQ1 Raw	14	GP15 STS1	0	Only valid for pins configured as GPIO input (does not include
Status_17	13	GP14 STS1	0	DSPGPIO inputs).
	12	GP13_STS1	0	
	11	GP12 STS1	0	
	10	GP11 STS1	0	
	9	GP10 STS1	0	
	8	GP9 STS1	0	
	7	GP8 STS1	0	
	6	GP7 STS1	0	
	5	GP6_STS1	0	
	4	GP5 STS1	0	
	3	GP4 STS1	0	
	2	GP3 STS1	0	
·	1	GP2 STS1	0	
	0	GP1 STS1	0	
R6293 (0x1895)	0	EVENT1_NOT_EMPTY_STS1	0	Event Log 1 FIFO Not-Empty status
IRQ1_Raw_				0 = FIFO Empty
Status 22				1 = FIFO Not Empty
	0	EVENT1_FULL_STS1	0	Event Log 1 FIFO Full status
IRQ1_Raw_				0 = FIFO Not Full
Status 23				1 = FIFO Full
R6295 (0x1897)	0	EVENT1 WMARK STS1	0	Event Log 1 FIFO Watermark status
IRQ1 Raw				0 = FIFO Watermark not reached
Status 24				1 = FIFO Watermark reached
R6296 (0x1898)	0	DSP1 DMA STS1	0	DSP1 DMA status
IRQ1_Raw_			-	0 = Normal
Status_25			1	1 = All enabled WDMA buffers filled, and all enabled RDMA
510103_20				buffers emptied
R6301 (0x189D)	0	DSP1_BUSY_STS1	0	DSP1 Busy status
IRQ1_Raw_			1	0 = DSP Idle
Status_30				1 = DSP Busy



Register Address	Bit	Label	Default	Description
R6307 (0x18A3)	3	TIMER_ALM1_CH4_STS1	0	Alarm 1 Channel 4 status
IRQ1_Raw_				0 = Alarm idle
Status_36				1 = Alarm output asserted
	2	TIMER_ALM1_CH3_STS1	0	Alarm 1 Channel 3 status
				0 = Alarm idle
				1 = Alarm output asserted
	1	TIMER_ALM1_CH2_STS1	0	Alarm 1 Channel 2 status
				0 = Alarm idle
				1 = Alarm output asserted
	0	TIMER_ALM1_CH1_STS1	0	Alarm 1 Channel 1 status
				0 = Alarm idle
				1 = Alarm output asserted

1. The BOOT DONE EINT1 interrupt is 0 (unmasked) by default; all other interrupts are 1 (masked) by default.

The IRQ2 interrupt, mask, and status control registers are described in Table 4-100.

Register Address Bit Label Default Description CTRLIF_ERR_EINT2 R6400 (0x1900) 12 0 Control Interface Error Interrupt (Rising edge triggered) IRQ2_Status_1 Note: Cleared when a 1 is written. 9 SYSCLK_FAIL_EINT2 0 SYSCLK Fail Interrupt (Rising edge triggered) Note: Cleared when a 1 is written. BOOT DONE EINT2 0 Boot Done Interrupt (Rising edge triggered) 7 Note: Cleared when a 1 is written. R6401 (0x1901) 14 DSPCLK ERR EINT2 0 DSPCLK Error Interrupt (Rising edge triggered) IRQ2 Status 2 Note: Cleared when a 1 is written. 0 13 ASYNCCLK ERR EINT2 ASYNCCLK Error Interrupt (Rising edge triggered) Note: Cleared when a 1 is written. 12 SYSCLK ERR EINT2 0 SYSCLK Error Interrupt (Rising edge triggered) Note: Cleared when a 1 is written. 9 FLL2_LOCK_EINT2 0 FLL2 Lock Interrupt (Rising and falling edge triggered) Note: Cleared when a 1 is written. FLL1_LOCK_EINT2 0 FLL1 Lock Interrupt (Rising and falling edge triggered) 8 Note: Cleared when a 1 is written. FLL2 Reference Lost Interrupt (Rising edge triggered) 6 FLL2 REF LOST EINT2 0 Note: Cleared when a 1 is written. 0 FLL1 Reference Lost Interrupt (Rising edge triggered) 5 FLL1 REF LOST EINT2 Note: Cleared when a 1 is written. R6405 (0x1905) 9 MICDET2 EINT2 0 Mic/Accessory Detect 2 Interrupt (Detection event triggered) Note: Cleared when a 1 is written. IRQ2 Status 6 8 MICDET1_EINT2 0 Mic/Accessory Detect 1 Interrupt (Detection event triggered) Note: Cleared when a 1 is written. 0 HPDET EINT2 0 Headphone Detect Interrupt (Rising edge triggered)

Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6406 (0x1906)	3	JD3_FALL_EINT2	0	JD3 Interrupt (Falling edge triggered)
IRQ2_Status_7				Note: Cleared when a 1 is written.
	2	JD3_RISE_EINT2	0	JD3 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	5	MICD_CLAMP_FALL_EINT2	0	MICDET Clamp Interrupt (Falling edge triggered)
				Indicates a falling edge transition on MICDET Clamp 1 or MICDET Clamp 2. Separate x_EINT2 bits are not provided—it is assumed that a maximum of one clamp is active at any time.
				Note: Cleared when a 1 is written.
	4	MICD_CLAMP_RISE_EINT2	0	MICDET Clamp Interrupt (Rising edge triggered)
				Indicates a rising edge transition on MICDET Clamp 1 or MICDET Clamp 2. Separate x_EINT2 bits are not provided—it is assumed that a maximum of one clamp is active at any time.
				Note: Cleared when a 1 is written.
	3	JD2_FALL_EINT2	0	JD2 Interrupt (Falling edge triggered)
				Note: Cleared when a 1 is written.
	2	JD2_RISE_EINT2	0	JD2 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	JD1_FALL_EINT2	0	JD1 Interrupt (Falling edge triggered)
				Note: Cleared when a 1 is written.
	0	JD1_RISE_EINT2	0	JD1 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6408 (0x1908) IRQ2 Status 9	9	ASRC1_IN2_LOCK_EINT2	0	ASRC1 IN2 Lock Interrupt (Rising and falling edge triggered) Note: Cleared when a 1 is written.
	8	ASRC1 IN1 LOCK EINT2	0	ASRC1 IN1 Lock Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	2	INPUTS_SIG_DET_EINT2	0	Input Path Signal-Detect Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	1	DRC2_SIG_DET_EINT2	0	DRC2 Signal-Detect Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	0	DRC1_SIG_DET_EINT2	0	DRC1 Signal-Detect Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6410 (0x190A)	15	DSP_IRQ16_EINT2	0	DSP IRQ16 Interrupt (Rising edge triggered)
IRQ2_Status_11				Note: Cleared when a 1 is written.
	14	DSP_IRQ15_EINT2	0	DSP IRQ15 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	13	DSP_IRQ14_EINT2	0	DSP IRQ14 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	12	DSP_IRQ13_EINT2	0	DSP IRQ13 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	11	DSP_IRQ12_EINT2	0	DSP IRQ12 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	10	DSP_IRQ11_EINT2	0	DSP IRQ11 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	9	DSP_IRQ10_EINT2	0	DSP IRQ10 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	8	DSP_IRQ9_EINT2	0	DSP IRQ9 Interrupt (Rising edge triggered)
			-	Note: Cleared when a 1 is written.
	7	DSP_IRQ8_EINT2	0	DSP IRQ8 Interrupt (Rising edge triggered)
			-	Note: Cleared when a 1 is written.
	6	DSP_IRQ7_EINT2	0	DSP IRQ7 Interrupt (Rising edge triggered)
		_ ~ _	-	Note: Cleared when a 1 is written.
	5	DSP IRQ6 EINT2	0	DSP IRQ6 Interrupt (Rising edge triggered)
	•		C C	Note: Cleared when a 1 is written.
	4	DSP IRQ5 EINT2	0	DSP IRQ5 Interrupt (Rising edge triggered)
	-			Note: Cleared when a 1 is written.
	3	DSP IRQ4 EINT2	0	DSP IRQ4 Interrupt (Rising edge triggered)
	•		C C	Note: Cleared when a 1 is written.
	2	DSP_IRQ3_EINT2	0	DSP IRQ3 Interrupt (Rising edge triggered)
	_		C C	Note: Cleared when a 1 is written.
	1	DSP IRQ2 EINT2	0	DSP IRQ2 Interrupt (Rising edge triggered)
	-			Note: Cleared when a 1 is written.
	0	DSP_IRQ1_EINT2	0	DSP IRQ1 Interrupt (Rising edge triggered)
	Ŭ		Ũ	Note: Cleared when a 1 is written.
R6411 (0x190B)	9	HP4R_SC_EINT2	0	HPOUT4R Short Circuit Interrupt (Rising edge triggered)
IRQ2_Status_12	Ū.		°,	Note: Cleared when a 1 is written.
	8	HP4L_SC_EINT2	0	HPOUT4L Short Circuit Interrupt (Rising edge triggered)
	•	····	°,	Note: Cleared when a 1 is written.
	5	HP3R SC EINT2	0	HPOUT3R Short Circuit Interrupt (Rising edge triggered)
	Ŭ		Ũ	Note: Cleared when a 1 is written.
	4	HP3L_SC_EINT2	0	HPOUT3L Short Circuit Interrupt (Rising edge triggered)
	•		Ũ	Note: Cleared when a 1 is written.
	3	HP2R_SC_EINT2	0	HPOUT2R Short Circuit Interrupt (Rising edge triggered)
	5		Ŭ	Note: Cleared when a 1 is written.
	2	HP2L SC EINT2	0	HPOUT2L Short Circuit Interrupt (Rising edge triggered)
	2		U	Note: Cleared when a 1 is written.
	1	HP1R_SC_EINT2	0	HPOUT1R Short Circuit Interrupt (Rising edge triggered)
	I		0	Note: Cleared when a 1 is written.
	0	HP1L_SC_EINT2	0	HPOUT1L Short Circuit Interrupt (Rising edge triggered)
	0		0	Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6412 (0x190C)	5	HP3R_ENABLE_DONE_EINT2	0	HPOUT3R/HPOUT4R Enable Interrupt (Rising edge triggered)
IRQ2_Status_13				Note: Cleared when a 1 is written.
	4	HP3L_ENABLE_DONE_EINT2	0	HPOUT3L/HPOUT4L Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	HP2R_ENABLE_DONE_EINT2	0	HPOUT2R Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	2	HP2L_ENABLE_DONE_EINT2	0	HPOUT2L Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	HP1R_ENABLE_DONE_EINT2	0	HPOUT1R Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	HP1L_ENABLE_DONE_EINT2	0	HPOUT1L Enable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6413 (0x190D)	5	HP3R_DISABLE_DONE_EINT2	0	HPOUT3R/HPOUT4R Disable Interrupt (Rising edge triggered)
IRQ2_Status_14				Note: Cleared when a 1 is written.
	4	HP3L_DISABLE_DONE_EINT2	0	HPOUT3L/HPOUT4L Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	3	HP2R_DISABLE_DONE_EINT2	0	HPOUT2R Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	2	HP2L_DISABLE_DONE_EINT2	0	HPOUT2L Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	HP1R_DISABLE_DONE_EINT2	0	HPOUT1R Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	HP1L_DISABLE_DONE_EINT2	0	HPOUT1L Disable Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6414 (0x190E)	12	DFC_SATURATE_EINT2	0	DFC Saturate Interrupt (Rising edge triggered)
IRQ2_Status_15				Note: Cleared when a 1 is written.



Register Address	Bit	Label	Default	Description
R6416 (0x1910)	15	GP16_EINT2	0	GPIO16 Interrupt (Rising and falling edge triggered)
IRQ2_Status_17				Note: Cleared when a 1 is written.
	14	GP15_EINT2	0	GPIO15 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	13	GP14_EINT2	0	GPIO14 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	12	GP13_EINT2	0	GPIO13 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	11	GP12_EINT2	0	GPIO12 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	10	GP11_EINT2	0	GPIO11 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	9	GP10_EINT2	0	GPIO10 Interrupt (Rising and falling edge triggered)
		_		Note: Cleared when a 1 is written.
	8	GP9_EINT2	0	GPIO9 Interrupt (Rising and falling edge triggered)
	-			Note: Cleared when a 1 is written.
	7	GP8 EINT2	0	GPIO8 Interrupt (Rising and falling edge triggered)
				Note: Cleared when a 1 is written.
	6	GP7 EINT2	0	GPIO7 Interrupt (Rising and falling edge triggered)
	-			Note: Cleared when a 1 is written.
	5	GP6_EINT2	0	GPIO6 Interrupt (Rising and falling edge triggered)
	Ŭ		Ũ	Note: Cleared when a 1 is written.
	4	GP5 EINT2	0	GPIO5 Interrupt (Rising and falling edge triggered)
			Ũ	Note: Cleared when a 1 is written.
	3	GP4 EINT2	0	GPIO4 Interrupt (Rising and falling edge triggered)
	Ŭ		Ū	Note: Cleared when a 1 is written.
	2	GP3 EINT2	0	GPIO3 Interrupt (Rising and falling edge triggered)
	~		Ū	Note: Cleared when a 1 is written.
	1	GP2 EINT2	0	GPIO2 Interrupt (Rising and falling edge triggered)
			Ũ	Note: Cleared when a 1 is written.
	0	GP1_EINT2	0	GPIO1 Interrupt (Rising and falling edge triggered)
	Ŭ		Ū	Note: Cleared when a 1 is written.
R6420 (0x1914)	0	TIMER1_EINT2	0	Timer 1 Interrupt (Rising edge triggered)
. ,	Ŭ		Ū	Note: Cleared when a 1 is written.
IRQ2_Status_21 R6421 (0x1915)	0	EVENT1_NOT_EMPTY_EINT2	0	Event Log 1 FIFO Not Empty Interrupt (Rising edge triggered)
· · ·	0	EVENTI_NOT_EMPTT_EINT2	0	Note: Cleared when a 1 is written.
IRQ2_Status_22			0	
R6422 (0x1916)	0	EVENT1_FULL_EINT2	0	Event Log 1 FIFO Full Interrupt (Rising edge triggered)
IRQ2_Status_23	<u> </u>			Note: Cleared when a 1 is written.
R6423 (0x1917)	0	EVENT1_WMARK_EINT2	0	Event Log 1 FIFO Watermark Interrupt (Rising edge triggered)
IRQ2_Status_24	ļ			Note: Cleared when a 1 is written.
R6424 (0x1918)	0	DSP1_DMA_EINT2	0	DSP1 DMA Interrupt (Rising edge triggered)
IRQ2_Status_25				Note: Cleared when a 1 is written.
R6426 (0x191A)	0	DSP1_START1_EINT2	0	DSP1 Start 1 Interrupt (Rising edge triggered)
IRQ2_Status_27				Note: Cleared when a 1 is written.
R6427 (0x191B)	0	DSP1_START2_EINT2	0	DSP1 Start 2 Interrupt (Rising edge triggered)
IRQ2 Status 28				Note: Cleared when a 1 is written.
R6429 (0x191D)	0	DSP1 BUSY EINT2	0	DSP1 Busy Interrupt (Rising edge triggered)
IRQ2 Status 30	-		-	Note: Cleared when a 1 is written.
R6432 (0x1920)	0	DSP1_BUS_ERR_EINT2	0	DSP1 Bus Error Interrupt (Rising edge triggered)
IRQ2_Status_33	J		Ŭ	Note: Cleared when a 1 is written.
				note. Cleared when a no whiten.



Register Address	Bit	Label	Default	Description
R6435 (0x1923)	3	TIMER_ALM1_CH4_EINT2	0	Alarm 1 Channel 4 Interrupt (Rising edge triggered)
IRQ2_Status_36				Note: Cleared when a 1 is written.
	2	TIMER_ALM1_CH3_EINT2	0	Alarm 1 Channel 3 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	1	TIMER_ALM1_CH2_EINT2	0	Alarm 1 Channel 2 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
	0	TIMER_ALM1_CH1_EINT2	0	Alarm 1 Channel 1 Interrupt (Rising edge triggered)
				Note: Cleared when a 1 is written.
R6464 (0x1940) to		IM_*	1	For each x_EINT2 interrupt bit in R6400 to R6435, a corresponding mask bit (IM_*) is provided in R6464 to R6499.
R6499 (0x1963)				The mask bits are coded as follows:
· · · ·				0 = Do not mask interrupt
				1 = Mask interrupt
R6528 (0x1980)	12	CTRLIF_ERR_STS2	0	Control Interface Error Status
IRQ2_Raw_				0 = Normal
Status_1				1 = Control Interface Error
	7	BOOT_DONE_STS2	0	Boot Status
				0 = Busy (boot sequence in progress)
				1 = Idle (boot sequence completed)
				Control register writes should not be attempted until Boot Sequence has completed.
R6529 (0x1981)	14	DSPCLK_ERR_STS2	0	DSPCLK Error Interrupt Status
IRQ2_Raw_				0 = Normal
Status_2				1 = Insufficient DSPCLK cycles for one or more of the requested DSP1 clock frequencies
	13	ASYNCCLK_ERR_STS2	0	ASYNCCLK Error Interrupt Status
				0 = Normal
				1 = Insufficient ASYNCCLK cycles for the requested signal path functionality
	12	SYSCLK_ERR_STS2	0	SYSCLK Error Interrupt Status
				0 = Normal
				1 = Insufficient SYSCLK cycles for the requested signal path functionality
	9	FLL2 LOCK STS2	0	FLL2 Lock Status
				0 = Not locked
				1 = Locked
	8	FLL1 LOCK STS2	0	FLL1 Lock Status
				0 = Not locked
				1 = Locked
	6	FLL2_REF_LOST_STS2	0	FLL2 Reference Lost Status
				0 = Normal
				1 = Reference Lost
	5	FLL1_REF_LOST_STS2	0	FLL1 Reference Lost Status
				0 = Normal
				1 = Reference Lost



Register Address	Bit	Label	Default	Description
R6534 (0x1986)	8	JD3_STS2	0	JACKDET3 input status
IRQ2_Raw_				0 = Jack not detected
Status_7				1 = Jack is detected
				(Assumes the JACKDET3 pin is pulled low on jack insertion.)
	4	MICD_CLAMP_STS2	0	MICDET Clamp status
				0 = Clamp disabled
				1 = Clamp enabled
				0 = Clamp disabled
				1 = Clamp enabled
				Separate _STS bits are not provided for each clamp—it is
				assumed that a maximum of one clamp is active at any time. The clamp override condition (MICD_CLAMP <i>n</i> _OVD = 1) is not
	2		0	
	2	JD2_STS2	0	JACKDET2 input status
				0 = Jack not detected
				1 = Jack is detected
				(Assumes the JACKDET2 pin is pulled low on jack insertion.)
	0	JD1_STS2	0	JACKDET1 input status
				0 = Jack not detected
				1 = Jack is detected
			-	(Assumes the JACKDET1 pin is pulled low on jack insertion.)
R6536 (0x1988) IRQ2_Raw_ Status_9	9	ASRC1_IN2_LOCK_STS2	0	ASRC1 IN2 Lock Status
				0 = Not locked
				1 = Locked
	8	ASRC1_IN1_LOCK_STS2	0	ASRC1 IN1 Lock Status
				0 = Not locked
				1 = Locked
	2	INPUTS_SIG_DET_STS2	0	Input Path Signal-Detect Status
				0 = Normal
				1 = Signal detected
	1	DRC2_SIG_DET_STS2	0	DRC2 Signal-Detect Status
				0 = Normal
				1 = Signal detected
	0	DRC1_SIG_DET_STS2	0	DRC1 Signal-Detect Status
				0 = Normal
				1 = Signal detected



Register Address	Bit	Label	Default	Description
R6539 (0x198B)	9	HP4R_SC_STS2	0	HPOUT4R Short Circuit Status
IRQ2_Raw_				0 = Normal
Status_12				1 = Short Circuit detected
	8	HP4L_SC_STS2	0	HPOUT4L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	5	HP3R_SC_STS2	0	HPOUT3R Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	4	HP3L_SC_STS2	0	HPOUT3L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	3	HP2R_SC_STS2	0	HPOUT2R Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	2	HP2L_SC_STS2	0	HPOUT2L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	1	HP1R_SC_STS2	0	HPOUT1R Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
	0	HP1L_SC_STS2	0	HPOUT1L Short Circuit Status
				0 = Normal
				1 = Short Circuit detected
R6540 (0x198C)	5	HP3R ENABLE DONE STS2	0	HPOUT3R/HPOUT4R Enable Status
IRQ2 Raw				0 = Busy (sequence in progress)
Status_13				1 = Idle (sequence completed)
	4	HP3L_ENABLE_DONE_STS2	0	HPOUT3L/HPOUT4L Enable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	3	HP2R_ENABLE_DONE_STS2	0	HPOUT2R Enable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	2	HP2L_ENABLE_DONE_STS2	0	HPOUT2L Enable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	1	HP1R_ENABLE_DONE_STS2	0	HPOUT1R Enable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	0	HP1L ENABLE DONE STS2	0	HPOUT1L Enable Status
	-		-	0 = Busy (sequence in progress)
				1 = Idle (sequence completed)

Table 4-100. Interrupt 2 Control Registers (Cont.)



Register Address	Bit	Label	Default	Description
R6541 (0x198D)	5	HP3R_DISABLE_DONE_STS2	0	HPOUT3R/HPOUT4R Disable Status
IRQ2_Raw_				0 = Busy (sequence in progress)
Status_14				1 = Idle (sequence completed)
	4	HP3L_DISABLE_DONE_STS2	0	HPOUT3L/HPOUT4L Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	3	HP2R_DISABLE_DONE_STS2	0	HPOUT2R Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	2	HP2L_DISABLE_DONE_STS2	0	HPOUT2L Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	1	HP1R_DISABLE_DONE_STS2	0	HPOUT1R Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
	0	HP1L_DISABLE_DONE_STS2	0	HPOUT1L Disable Status
				0 = Busy (sequence in progress)
				1 = Idle (sequence completed)
R6544 (0x1990)	15	GP16_STS2	0	GPIO <i>n</i> Input status
IRQ2_Raw_	14	GP15_STS2	0	Reads back the logic level of GPIO <i>n</i> .
Status_17	13	GP14_STS2	0	Only valid for pins configured as GPIO input (does not include
	12	GP13_STS2	0	DSPGPIO inputs).
	11	GP12_STS2	0	
	10	GP11_STS2	0	
	9	GP10_STS2	0	
	8	GP9_STS2	0	
	7	GP8_STS2	0	-
	6	GP7_STS2	0	-
	5	GP6_STS2	0	-
	4	GP5_STS2	0	-
	3	GP4_STS2	0	-
	2	GP3_STS2 GP2_STS2	0	-
	0	GP2_5152 GP1 STS2	0	-
R6549 (0x1995)	0	EVENT1_NOT_EMPTY_STS2	0	Event Log 1 FIFO Not-Empty status
· · ·	0		0	0 = FIFO Empty
IRQ2_Raw_ Status 22				1 = FIFO Not Empty
R6550 (0x1996)	0	EVENT1_FULL_STS2	0	Event Log <i>n</i> FIFO Full status
	0		0	0 = FIFO Not Full
IRQ2_Raw_ Status 23				1 = FIFO Full
R6551 (0x1997)	0	EVENT1_WMARK_STS2	0	Event Log 1 FIFO Watermark status
	Ū		Ũ	0 = FIFO Watermark not reached
IRQ2_Raw_ Status 24				1 = FIFO Watermark reached
R6552 (0x1998)	0	DSP1 DMA STS2	0	DSP1 DMA status
	0			0 = Normal
IRQ2_Raw_				1 = All enabled WDMA buffers filled, and all enabled RDMA
Status_25				buffers emptied
R6557 (0x199D)	0	DSP1_BUSY_STS2	0	DSP1 Busy status
IRQ2_Raw_				0 = DSP Idle
Status_30				1 = DSP Busy

Table 4-100. Interrupt 2 Control Registers (Cont.)



Register Address	Bit	Label	Default	Description
R6563 (0x19A3)	3	TIMER_ALM1_CH4_STS2	0	Alarm 1 Channel 4 status
IRQ2_Raw_				0 = Alarm idle
Status_36				1 = Alarm output asserted
	2	TIMER_ALM1_CH3_STS2	0	Alarm 1 Channel 3 status
				0 = Alarm idle
				1 = Alarm output asserted
	1	TIMER_ALM1_CH2_STS2	0	Alarm 1 Channel 2 status
				0 = Alarm idle
				1 = Alarm output asserted
	0	TIMER_ALM1_CH1_STS2	0	Alarm 1 Channel 1 status
				0 = Alarm idle
				1 = Alarm output asserted

Table 4-100. Interrupt 2 Control Registers (Cont.)

The IRQ output and polarity control registers are described in Table 4-101.

Register Address	Bit	Label	Default	Description
R6784 (0x1A80)	11	IM_IRQ1	0	IRQ1 Output Interrupt mask.
IRQ1_CTRL				0 = Do not mask interrupt.
				1 = Mask interrupt.
	10	IRQ_POL	1	IRQ Output Polarity Select
				0 = Noninverted (Active High)
				1 = Inverted (Active Low)
	9	IRQ_OP_CFG	0	IRQ Output Configuration
				0 = CMOS
				1 = Open drain
R6786 (0x1A82)	11	IM_IRQ2	0	IRQ2 Output Interrupt mask.
IRQ2_CTRL				0 = Do not mask interrupt.
				1 = Mask interrupt.
R6816 (0x1AA0)	1	IRQ2_STS	0	IRQ2 Status. IRQ2_STS is the logical OR of all unmasked x_EINT2 interrupts.
Interrupt_Raw_				0 = Not asserted
Status_1				1 = Asserted
	0	IRQ1_STS	0	IRQ1 Status. IRQ1_STS is the logical OR of all unmasked x_EINT1 interrupts.
				0 = Not asserted
				1 = Asserted

Table 4-101. Interrupt Control Registers

4.16 Clocking and Sample Rates

The CS42L92 requires a clock reference for its internal functions and also for the input (ADC) paths, output (DAC) paths, and digital audio interfaces. Under typical clocking configurations, all commonly used audio sample rates can be derived directly from the external reference; for additional flexibility, the CS42L92 incorporates two FLL circuits to perform frequency conversion and filtering.

External clock signals may be connected via the MCLK1, MCLK2, and MCLK3 input pins. In AIF Slave Modes, the BCLK signals may be used as a reference for the system clocks. The SLIMbus interface can provide the clock reference, when used as the input to one of the FLLs. To avoid audible glitches, all clock configurations must be set up before enabling playback.

4.16.1 System Clocking Overview

The CS42L92 supports three primary clock domains—SYSCLK, ASYNCCLK, and DSPCLK.

The SYSCLK and ASYNCCLK clock domains are the reference clocks for all the audio signal paths on the CS42L92. Up to five different sample rates may be independently selected for specific audio interfaces and other input/output signal paths; each selected sample rate must be synchronized either to SYSCLK or to ASYNCCLK, as described in Section 4.16.2.



The SYSCLK and ASYNCCLK clock domains are independent (i.e., not synchronized). Stereo full-duplex sample-rate conversion is supported, allowing asynchronous audio data to be mixed and to be routed between independent interfaces. See Section 4.3 for further details.

The DSPCLK clock domain is the reference clock for the programmable DSP core on the CS42L92. A wide range of DSPCLK frequencies can be supported, and a programmable clock divider is provided for the DSP core, allowing the DSP clocking (and power consumption) to be optimized according to the applicable processing requirements. See Section 4.3 for further details.

Note that there is no requirement for DSPCLK to be synchronized to SYSCLK or ASYNCCLK. The DSPCLK controls the software execution in the DSP core; audio outputs from the DSP are synchronized either to SYSCLK or ASYNCCLK, regardless of the applicable DSPCLK rate.

Excluding the DSP core, each subsystem within the CS42L92 digital core is clocked at a dynamically controlled rate, limited by the SYSCLK (or ASYNCCLK) frequency, as applicable. For maximum signal mixing and processing capacity, it is recommended that the highest possible SYSCLK and ASYNCCLK frequencies are configured.

The DSP core is clocked at the DSPCLK rate (or supported divisions of the DSPCLK frequency). The DSPCLK configuration must ensure that sufficient clock cycles are available for the processing requirements of the DSP core. The requirements vary, according to the particular software that is in use.

4.16.2 Sample-Rate Control

The CS42L92 supports two independent clock domains for the audio signal paths, referenced to SYSCLK and ASYNCCLK respectively.

Different sample rates may be selected for each of the audio interfaces (AIF1, AIF2, AIF3, SLIMbus), and for the input (ADC) and output (DAC) paths. Each of these must be referenced either to SYSCLK or to ASYNCCLK. (Note that the SLIMbus interface supports multiple sample rates, selected independently for each input or output channel.)

Up to three different sample rates can be selected using SAMPLE_RATE_1, SAMPLE_RATE_2 and SAMPLE_RATE_3. These must each be numerically related to each other and to the SYSCLK frequency (further details of these requirements are provided in Table 4-102 and the accompanying text).

The remaining two sample rates can be selected using ASYNC_SAMPLE_RATE_1 and ASYNC_SAMPLE_RATE_2. These sample rates must be numerically related to each other and to the ASYNCCLK frequency (further details of these requirements are provided in Table 4-103 and the accompanying text),

Each of the audio interfaces, input paths, and output paths is associated with one of the sample rates selected by the SAMPLE_RATE_n or ASYNC_SAMPLE_RATE_n fields.

Note that if any two interfaces are operating at the same sample rate, but are not synchronized, one of these must be referenced to the ASYNCCLK domain, and the other to the SYSCLK domain.

When any of the SAMPLE_RATE_*n* or ASYNC_SAMPLE_RATE_*n* fields is written to, the activation of the new setting is automatically synchronized by the CS42L92 to ensure continuity of all active signal paths. The SAMPLE_RATE_*n*_STS and ASYNC_SAMPLE_RATE_*n*_STS bits provide indication of the sample rate selections that have been implemented.



The following restrictions must be observed regarding the sample-rate control configuration:

- Unless otherwise noted, the sample rate selection for all functions is valid from 8-192 kHz.
- The input (ADC/DMIC) signal paths must always be associated with the SYSCLK clocking domain.
- If 384- or 768-kHz DMICCLK clock rate is selected, the supported sample rate for the respective input paths is restricted as described in Table 4-1. The sample rate for the input signal paths can be set globally, or can be configured independently for each input channel—see Section 4.2.5.
- The S/PDIF sample rate is valid from 32–192 kHz. The output (DAC), digital audio interface (AIF), DSP core, and SLIMbus input/output sample rates are valid from 8–384 kHz.
- The asynchronous sample-rate converter (ASRC) supports sample rates 8–192 kHz. The ratio of the two sample rates must not exceed 6.
- The isochronous sample-rate converters (ISRCs) support sample rates 8–384 kHz. For each ISRC, the ratio of the applicable SAMPLE_RATE_*n* or ASYNC_SAMPLE_RATE_*n* fields must not exceed 24. The sample-rate conversion ratio must be an integer (1–24) or equal to 1.5.
- All external clock references (MCLK input or Slave Mode AIF input) must be within 1% of the applicable register field settings.

4.16.3 Automatic Sample-Rate Detection

The CS42L92 supports automatic sample-rate detection on the digital audio interfaces (AIF1–AIF3). Note that this is only possible when the respective interface is operating in Slave Mode (i.e., when LRCLK and BCLK are inputs to the CS42L92).

Automatic sample-rate detection is enabled by setting RATE_EST_ENA. The LRCLK input pin selected for sample-rate detection is set using LRCLK_SRC.

As many as four audio sample rates can be configured for automatic detection; these sample rates are selected using the SAMPLE_RATE_DETECT_*n* fields. Note that the function only detects sample rates that match one of the SAMPLE_RATE_DETECT_*n* fields.

If one of the selected audio sample rates is detected on the selected LRCLK input, the control-write sequencer is triggered. A unique sequence of actions may be programmed for each detected sample rate. Note that the applicable control sequences must be programmed by the user for each detection outcome; see Section 4.18.

The TRIG_ON_STARTUP bit controls whether the sample-rate detection circuit responds to the initial detection of the applicable interface (i.e., when the AIF*n* interface starts up).

- If TRIG_ON_STARTUP = 0, the detection circuit only responds (i.e., trigger the control-write sequencer) to a
 change in the detected sample rate—the initial sample-rate detection is ignored. (Note that the initial sample-rate
 detection is the first detection of a sample rate that matches one of the SAMPLE_RATE_DETECT_n fields.)
- If TRIG_ON_STARTUP = 1, the detection circuit triggers the control-write sequencer whenever a selected sample rate is detected, including when the AIF interface starts up, or when the sample-rate detection is first enabled.

As described above, setting TRIG_ON_STARTUP = 0 is designed to inhibit any response to the initial detection of a sample rate that matches one of the SAMPLE_RATE_DETECT_*n* fields. Note that, if the LRCLK_SRC setting is changed, or if the detection function is disabled and reenabled, a subsequent detection of a matching sample rate may trigger the control-write sequencer, regardless of the TRIG_ON_STARTUP setting.

There are some restrictions to be observed regarding the automatic sample-rate detection configuration, as noted in the following:

- The same sample rate must not be selected on more than one of the SAMPLE_RATE_DETECT_*n* fields.
- Sample rates 384 kHz and 352.8 kHz must not be selected concurrently.
- Sample rates 192 kHz and 176.4 kHz must not be selected concurrently.
- Sample rates 96 kHz and 88.2 kHz must not be selected concurrently.

The control registers associated with the automatic sample-rate detection function are described in Table 4-104.



4.16.4 System Clock Configuration

The system clocks (SYSCLK, ASYNCCLK and DSPCLK) may be provided directly from external inputs (MCLK, or Slave Mode BCLK inputs). Alternatively, these clocks can be derived using the integrated FLLs, with MCLK, BCLK, LRCLK or SLIMCLK as a reference. Each clock is configured independently, as described in the following sections.

The SYSCLK (and ASYNCCLK, when applicable) clocks must be configured and enabled before any audio path is enabled. The DSPCLK clock must be configured and enabled, if running firmware applications on the DSP core.

4.16.4.1 SYSCLK Configuration

The required SYSCLK frequency is dependent on the SAMPLE_RATE_*n* fields. Table 4-102 illustrates the valid SYSCLK frequencies for every supported sample rate.

The SYSCLK frequency must be valid for all of the SAMPLE_RATE_*n* fields. It follows that all of the SAMPLE_RATE_*n* fields must select numerically-related values, that is, all from the same group of sample rates as represented in Table 4-102.

SYSCLK Frequency (MHz)	SYSCLK_FREQ	SYSCLK_FRAC	Sample Rate (kHz)	SAMPLE_RATE_n
6.144	000	0	12	0x01
12.288	001		24	0x02
24.576 49.152	010 011		48	0x03
98.304	100		96	0x04
			192	0x05
			384	0x06
			8	0x11
			16	0x12
			32	0x13
5.6448	000	1	11.025	0x09
11.2896	001		22.05	0x0A
22.5792 45.1584	010 011 100		44.1	0x0B
90.3168			88.2	0x0C
			176.4	0x0D
			352.8	0x0E

Table 4-102. SYSCLK Frequency Selection

Note: The SAMPLE_RATE_*n* fields must each be set to a value from the same group of sample rates, and from the same group as the SYSCLK frequency.

SYSCLK_SRC is used to select the SYSCLK source, as described in Table 4-104. The source may be MCLK*n*, AIF*n*BCLK, or FLL*n*. If an FLL circuit is selected as the source, the relevant FLL must be enabled and configured, as described in Section 4.16.8.

Notes: If FLL1 is selected as SYSCLK source, two different clock frequencies are available. Typical use cases should select a SYSCLK frequency equal to F_{FLL1} × 2 (i.e., in the range 90–100 MHz). A lower frequency selection, equal to F_{FLL1}, is provided to support low-power always-on use cases.

If FLL2 is selected as SYSCLK source, the SYSCLK frequency is F_{FLL2} × 2 (i.e., 90–100 MHz).

SYSCLK_FREQ and SYSCLK_FRAC must be set according to the frequency of the selected SYSCLK source.

The SYSCLK-referenced circuits within the digital core are clocked at a dynamically controlled rate this is limited by the SYSCLK frequency. For maximum signal mixing and processing capacity, the highest possible SYSCLK frequency should be used.

The SAMPLE_RATE_*n* fields are set according to the sample rates that are required by one or more of the CS42L92 audio interfaces. The CS42L92 supports sample rates ranging from 8–384 kHz. See Section 4.16.2 for further details of the supported sample rates for each of the digital-core functions.

The SYSCLK signal is enabled by setting SYSCLK_ENA. The applicable clock source (MCLK*n*, AIF*n*BCLK, or FLL*n*) must be enabled before setting SYSCLK_ENA. This bit should be cleared before stopping or removing the applicable clock source.



The CS42L92 supports seamless switching between clock sources. To change the SYSCLK configuration while SYSCLK is enabled, the SYSCLK_FRAC, SYSCLK_FREQ, and SYSCLK_SRC fields must be updated together in one register write operation. Note that, if changing the frequency only (not the source), SYSCLK_ENA should be cleared before the clock frequency is updated. The current SYSCLK frequency and source can be read from the SYSCLK_FREQ_STS and SYSCLK_SRC_STS fields respectively.

The CS42L92 performs automatic checks to confirm that the SYSCLK frequency is high enough to support the commanded signal paths and processing functions. If the frequency is too low, an attempt to enable a signal path or processing function fails. Note that active signal paths are not affected under such circumstances.

The SYSCLK frequency check provides input to the interrupt-control circuit and can be used to trigger an interrupt event if the frequency is not high enough to support the commanded functionality; see Section 4.15.

4.16.4.2 ASYNCCLK Configuration

The required ASYNCCLK frequency is dependent on the ASYNC_SAMPLE_RATE_*n* fields. Table 4-103 illustrates the valid ASYNCCLK frequencies for every supported sample rate.

Note that, if all the sample rates in the system are synchronized to SYSCLK, the ASYNCCLK should be disabled (see Table 4-104). The associated register field values are not important in this case.

ASYNCCLK Frequency (MHz)	ASYNC_CLK_FREQ	Sample Rate (kHz)	ASYNC_SAMPLE_RATE_n
6.144	000	12	0x01
12.288	001	24	0x02
24.576 49.152	010 011	48	0x03
98.304	100	96	0x04
50.004	100	192	0x05
		384	0x06
		8	0x11
		16	0x12
		32	0x13
5.6448	000	11.025	0x09
11.2896	001	22.05	0x0A
22.5792 45.1584	010 011 100	44.1	0x0B
45.1584 90.3168		88.2	0x0C
00.0100	100	176.4	0x0D
		352.8	0x0E

Table 4-103.	ASYNCCLK Frequency Selection
--------------	-------------------------------------

Note: The ASYNC_SAMPLE_RATE_*n* fields must each be set to a value from the same group of sample rates, and from the same group as the ASYNCCLK frequency.

ASYNC_CLK_SRC is used to select the ASYNCCLK source, as described in Table 4-104. The source may be MCLK*n*, AIF*n*BCLK, or FLL*n*. If an FLL circuit is selected as the source, the relevant FLL must be enabled and configured, as described in Section 4.16.8.

Notes: If FLL1 is selected as ASYNCCLK source, two different clock frequencies are available. Typical use cases should select a SYSCLK frequency equal to F_{FLL1} × 2 (i.e., in the range 90–100 MHz). A lower frequency selection, equal to F_{FLL1}, is provided to support low-power always-on use cases.

If FLL2 is selected as ASYNCCLK source, the ASYNCCLK frequency is $F_{FLL2} \times 2$ (i.e., 90–100 MHz).

ASYNC_CLK_FREQ is set according to the frequency of the selected ASYNCCLK source.

The ASYNCCLK-referenced circuits within the digital core are clocked at a dynamically controlled rate that is limited by the ASYNCCLK frequency. For maximum signal mixing and processing capacity, the highest possible ASYNCCLK frequency should be used.

The ASYNC_SAMPLE_RATE_n fields are set according to the sample rates of any audio interface that is not synchronized to the SYSCLK clock domain.



The ASYNCCLK signal is enabled by setting ASYNC_CLK_ENA. The applicable clock source (MCLK*n*, AIF*n*BCLK, or FLL*n*) must be enabled before setting ASYNC_CLK_ENA. This bit should be cleared before stopping or removing the applicable clock source.

The CS42L92 supports seamless switching between clock sources. To change the ASYNCCLK configuration while ASYNCCLK is enabled, the ASYNC_CLK_FREQ and ASYNC_CLK_SRC fields must be updated together in one register write operation. Note that, if changing the frequency only (not the source), ASYNC_CLK_ENA should be cleared before the clock frequency is updated. The current ASYNCCLK frequency and source can be read from the ASYNC_CLK_FREQ_STS and ASYNC_CLK_SRC_STS fields respectively.

The CS42L92 performs automatic checks to confirm that the ASYNCCLK frequency is high enough to support the commanded signal paths and processing functions. If the frequency is too low, an attempt to enable a signal path or processing function fails. Note that active signal paths are not affected under such circumstances.

The ASYNCCLK frequency check provides input to the interrupt-control circuit and can be used to trigger an interrupt event if the frequency is not high enough to support the commanded functionality; see Section 4.15.

4.16.4.3 DSPCLK Configuration

The required DSPCLK frequency depends on the requirements of firmware loaded on the DSP core. The DSP is clocked at the DSPCLK rate or at supported divisions of the DSPCLK frequency; the DSPCLK configuration must ensure that sufficient clock cycles are available for the processing requirements. The requirements vary, according to the particular firmware that is in use.

A configurable clock divider is provided for the DSP core, allowing the DSP clocking (and power consumption) to be optimized according to the applicable processing requirements; see Section 4.4 for details.

DSP_CLK_FREQ must be configured for the applicable DSPCLK frequency. This field is coded in LSB units of 1/64 MHz. Note that, if the field coding cannot represent the DSPCLK frequency exactly, the DSPCLK frequency must be rounded down in the DSP_CLK_FREQ field.

The suggested method for calculating DSP_CLK_FREQ is to multiply the DSPCLK frequency by 64, round down to the nearest integer, and use the resulting integer as DSP_CLK_FREQ (LSB = 1).

DSP_CLK_SRC is used to select the DSPCLK source, as described in Table 4-104. The source may be MCLK*n*, AIF*n*BCLK, or FLL*n*. If an FLL circuit is selected as the source, the relevant FLL must be enabled and configured, as described in Section 4.16.8.

Notes: If FLL1 is selected as DSPCLK source, two different clock frequencies are available. Typical use cases should select a DSPCLK frequency equal to F_{FLL1} × 3 (i.e., in the range 135–150 MHz). A lower frequency selection, equal to F_{FLL1}, is provided to support low-power always-on use cases.

If FLL2 is selected as DSPCLK source, the DSPCLK frequency is F_{FLL2} × 3 (i.e., 135–150 MHz).

The DSPCLK signal is enabled by setting DSP_CLK_ENA. The applicable clock source (MCLK*n*, AIF*n*BCLK, or FLL*n*) must be enabled before setting DSP_CLK_ENA. This bit should be cleared when reconfiguring the clock sources.

The CS42L92 supports seamless switching between clock sources. To change the DSPCLK configuration while DSPCLK is enabled, the DSP_CLK_FREQ field must be updated before DSP_CLK_SRC. The new configuration becomes effective when the DSP_CLK_SRC field is written. Note that, if changing the frequency only (not the source), the DSP_CLK_ENA bit should be cleared before the clock frequency is updated. The current DSPCLK frequency and source can be read from the DSP_CLK_FREQ_STS and DSP_CLK_SRC_STS fields respectively.

In a typical application, DSPCLK and SYSCLK are derived from a single FLL source. Note that there is no requirement for DSPCLK to be synchronized to SYSCLK or ASYNCCLK. The DSPCLK controls the software execution in the DSP core; audio outputs from the DSP core are synchronized either to SYSCLK or ASYNCCLK, regardless of the applicable DSPCLK rate.



4.16.5 Miscellaneous Clock Controls

The CS42L92 incorporates a 32-kHz clock circuit, which is required for input signal debounce, microphone/accessory detect, and for the Charge Pump 2 (CP2) circuits. The 32-kHz clock must be configured and enabled whenever any of these features are used.

The 32-kHz clock can be generated automatically from SYSCLK, or may be provided externally via the MCLK1 or MCLK2 input pins. The 32-kHz clock source is selected using CLK_32K_SRC. The 32-kHz clock is enabled by setting CLK_32K_ENA.

A clock output (OPCLK) derived from SYSCLK can be output on a GPIO pin. A clock output (OPCLK_ASYNC) derived from ASYNCCLK can be output on a GPIO pin. See Section 4.14 for details on configuring a GPIO pin for these functions.

The CS42L92 provides integrated pull-down resistors on the MCLK1, MCLK2, and MCLK3 pins. This provides a flexible capability for interfacing with other devices.

The clocking scheme for the CS42L92 is shown in Fig. 4-65.



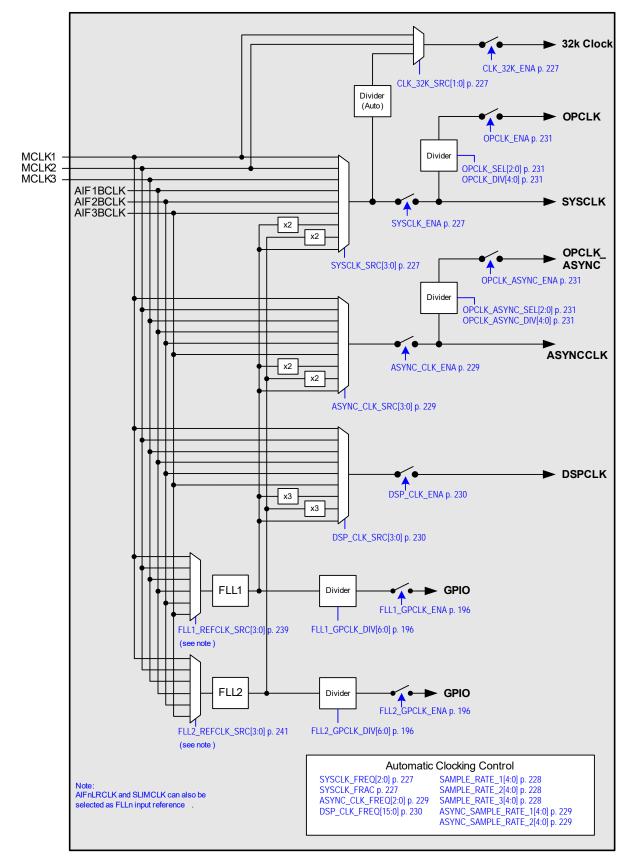


Figure 4-65. System Clocking



The CS42L92 clocking control registers are described in Table 4-104.

Table 4-104. Clocking Control

Register Address	Bit	Label	Default	Description
R256 (0x0100)	6	CLK_32K_ENA	0	32kHz Clock Enable
Clock_32k_1				0 = Disabled
				1 = Enabled
	1:0	CLK_32K_	10	32kHz Clock Source
		SRC[1:0]		00 = MCLK1 (direct)
				01 = MCLK2 (direct)
				10 = SYSCLK (automatically divided)
				11 = Reserved
R257 (0x0101)	15	SYSCLK_FRAC	0	SYSCLK Frequency
System_Clock_1				0 = SYSCLK is a multiple of 6.144MHz
				1 = SYSCLK is a multiple of 5.6448MHz
	10:8	SYSCLK_	100	SYSCLK Frequency
		FREQ[2:0]		000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				100 = 98.304 MHz (90.3168 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related sample rates only (i.e., SAMPLE_RATE_ <i>n</i> = 01XXX).
	6	SYSCLK_ENA	0	SYSCLK Control
				0 = Disabled
				1 = Enabled
				SYSCLK should only be enabled after the applicable clock source has been configured and enabled.
				Clear this bit before stopping the reference clock or changing the reference clock frequency. Note that the SYSCLK frequency can be changed without disabling, provided the clock source is also changed at the same time.
	3:0	SYSCLK_	0100	SYSCLK Source
		SRC[3:0]		0000 = MCLK1
				0001 = MCLK2
				0010 = MCLK3
				0100 = FLL1 × 2
				0101 = FLL2 × 2
				1000 = AIF1BCLK
				1001 = AIF2BCLK
				1010 = AIF3BCLK
				1111 = FLL1
				All other codes are reserved
		l	1	



Register Address	Bit	Label	Default	Description
R258 (0x0102)	4:0	SAMPLE_RATE_	0x11	Sample Rate 1 Select
Sample_rate_1		1[4:0]		0x00 = None
				0x01 = 12 kHz
				0x02 = 24 kHz
				0x03 = 48 kHz
				0x04 = 96 kHz
				0x05 = 192 kHz
				0x06 = 384 kHz
				0x09 = 11.025 kHz
				0x0A = 22.05 kHz
				0x0B = 44.1 kHz
				0x0C = 88.2 kHz
				0x0D = 176.4 kHz
				0x0E = 352.8 kHz
				0x11 = 8 kHz
				0x12 = 16 kHz
				0x13 = 32 kHz
				All other codes are reserved
R259 (0x0103)	4:0	SAMPLE_RATE_	0x11	Sample Rate 2 Select
Sample_rate_2		2[4:0]		Field coding is same as SAMPLE_RATE_1.
R260 (0x0104)	4:0	SAMPLE_RATE_	0x11	Sample Rate 3 Select
Sample_rate_3		3[4:0]		Field coding is same as SAMPLE_RATE_1.
R266 (0x010A)	4:0	SAMPLE_RATE_	0x00	Sample Rate 1 Status (Read only)
Sample_rate_1_ status		1_STS[4:0]		Field coding is same as SAMPLE_RATE_1.
R267 (0x010B)	4:0	SAMPLE_RATE_	0x00	Sample Rate 2 Status (Read only)
Sample_rate_2_ status		2_STS[4:0]		Field coding is same as SAMPLE_RATE_1.
R268 (0x010C)	4:0	SAMPLE_RATE_	0x00	Sample Rate 3 Status (Read only)
Sample_rate_3_ status		3_STS[4:0]		Field coding is same as SAMPLE_RATE_1.



Register Address	Bit	Label	Default	Description
R274 (0x0112)	10:8	ASYNC_CLK_	011	ASYNCCLK Frequency
Async_clock_1		FREQ[2:0]		000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				100 = 98.304 MHz (90.3168 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related sample rates only (i.e., ASYNC_SAMPLE_RATE_n = 01XXX).
	6	ASYNC_CLK_	0	ASYNCCLK Control
		ENA		0 = Disabled
				1 = Enabled
				ASYNCCLK should only be enabled after the applicable clock source has been configured and enabled.
				Clear this bit before stopping the reference clock or changing the reference clock frequency. Note that the ASYNCCLK frequency can be changed without disabling, provided the clock source is also changed at the same time.
	3:0	ASYNC_CLK_	0101	ASYNCCLK Source
		SRC[3:0]		0000 = MCLK1
				0001 = MCLK2
				0010 = MCLK3
				0100 = FLL1 × 2
				0101 = FLL2 × 2
				1000 = AIF1BCLK
				1001 = AIF2BCLK
				1010 = AIF3BCLK
				1111 = FLL1
				All other codes are reserved
R275 (0x0113)	4:0	ASYNC_	0x11	ASYNC Sample Rate 1 Select
Async_sample_		SAMPLE_RATE_		0x00 = None
rate_1		1[4:0]		0x01 = 12 kHz
				0x02 = 24 kHz
				0x03 = 48 kHz
				0x04 = 96 kHz
				0x05 = 192 kHz
				0x06 = 384 kHz
				0x09 = 11.025 kHz
				0x0A = 22.05 kHz
				0x0B = 44.1 kHz
				0x0C = 88.2 kHz
				0x0D = 176.4 kHz
				0x0E = 352.8 kHz
				0x11 = 8 kHz
				0x12 = 16 kHz
				0x13 = 32 kHz
D070 (0.0111)	4.0	A C)/() (C	0.44	All other codes are reserved
R276 (0x0114)	4:0	ASYNC_ SAMPLE_RATE_	0x11	ASYNC Sample Rate 2 Select
Async_sample_ rate_2		2[4:0]		Field coding is same as ASYNC_SAMPLE_RATE_1.
R283 (0x011B)	4:0	ASYNC_	0x00	ASYNC Sample Rate 1 Status (Read only)
Async_sample_		SAMPLE_RATE_		Field coding is same as ASYNC_SAMPLE_RATE_1.
rate_1_status		1_STS[4:0]		
R284 (0x011C)	4:0	ASYNC_	0x00	ASYNC Sample Rate 2 Status (Read only)
Async_sample_		SAMPLE_RATE_ 2_STS[4:0]		Field coding is same as ASYNC_SAMPLE_RATE_1.
rate_2_status		2_010[4.0]		



Register Address	Bit	Label	Default	Description
R288 (0x0120)	6	DSP_CLK_ENA	0	DSPCLK Control
DSP_Clock_1				0 = Disabled
				1 = Enabled
				DSPCLK should only be enabled after the applicable clock source has been
				configured and enabled.
				Clear this bit before stopping the reference clock or changing the reference clock
				frequency. Note that the DSPCLK frequency can be changed without disabling, provided the clock source is also changed at the same time.
	3:0	DSP CLK	0101	DSPCLK Source
	5.0	SRC[3:0]	0101	0000 = MCLK1
				0001 = MCLK2
				0010 = MCLK3
				$0100 = FLL1 \times 3$
				$0101 = FLL2 \times 3$
				1000 = AIF1BCLK
				1001 = AIF2BCLK
				1010 = AIF3BCLK
				1111 = FLL1
				All other codes are reserved
R290 (0x0122)	15:0	DSP CLK	0x0000	DSPCLK Frequency
DSP_Clock_2		FREQ[15:0]		Coded as LSB = 1/64 MHz, Valid from 5.6 MHz to 148 MHz.
				Note that, if this field is written while DSPCLK is enabled, the new frequency does
				not become effective until DSP_CLK_SRC is updated. To reconfigure DSPCLK
				while DSPCLK is enabled, the DSP_CLK_FREQ field must be updated before
				DSP_CLK_SRC.
R294 (0x0126)	15:0	DSP_CLK_	0x0000	DSPCLK Frequency (Read only)
DSP_Clock_4		FREQ_STS[15:0]		Coded as LSB = 1/64 MHz.
R295 (0x0127)	3:0	DSP_CLK_SRC_	0101	DSPCLK Source (Read only)
DSP_Clock_5		STS[3:0]		0000 = MCLK1
				0001 = MCLK2
				0010 = MCLK3
				0100 = FLL1 × 3
				0101 = FLL2 × 3
				1000 = AIF1BCLK
				1001 = AIF2BCLK
				1010 = AIF3BCLK
				1111 = FLL1
				All other codes are reserved



Register Address	Bit	Label	Default	Description
R329 (0x0149)	15	OPCLK_ENA	0	OPCLK Enable
Output_system_				0 = Disabled
clock				1 = Enabled
	7:3	OPCLK_DIV[4:0]	0x00	OPCLK Divider
				0x02 = Divide by 2
				0x04 = Divide by 4
				0x06 = Divide by 6
				(even numbers only)
				0x1E = Divide by 30
				Note that only even numbered divisions (2, 4, 6, etc.) are valid selections.
				All other codes are reserved when the OPCLK signal is enabled.
	2:0	OPCLK_SEL[2:0]	000	OPCLK Source Frequency
				000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related SYSCLK rates only (i.e.,
				SAMPLE_RATE_n = 01XXX).
				The OPCLK Source Frequency must be less than or equal to the SYSCLK frequency.
R330 (0x014A)	15	OPCLK_ASYNC_	0	OPCLK_ASYNC Enable
Output_async_		ENA		0 = Disabled
clock				1 = Enabled
	7:3	OPCLK_ASYNC_	0x00	OPCLK_ASYNC Divider
		DIV[4:0]		0x02 = Divide by 2
				0x04 = Divide by 4
				0x06 = Divide by 6
				(even numbers only)
				0x1E = Divide by 30
				Note that only even numbered divisions (2, 4, 6, etc.) are valid selections.
				All other codes are reserved when the OPCLK_ASYNC signal is enabled.
	2:0	OPCLK ASYNC	000	OPCLK ASYNC Source Frequency
		SEL[2:0]		000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related ASYNCCLK rates only (i.e., ASYNC_SAMPLE_RATE_n = 01XXX).
				The OPCLK ASYNC Source Frequency must be less than or equal to the
				ASYNCLK_ASYNC Source Frequency must be less than of equal to the
R334 (0x014E)	9	MCLK3_PD	0	MCLK3 Pull-Down Control
Clock Gen Pad	0		Ŭ	0 = Disabled
Ctrl				1 = Enabled
	8	MCLK2 PD	0	MCLK2 Pull-Down Control
	0		0	0 = Disabled
			0	1 = Enabled
	7	MCLK1_PD	0	MCLK1 Pull-Down Control
				0 = Disabled
				1 = Enabled



Register Address	Bit	Label	Default	Description			
R338 (0x0152)	4	TRIG_ON_	0	Automatic Sample-Rate Detection Start-Up select			
Rate_Estimator_1		STARTUP		0 = Do not trigger Write Sequencer on initial detection			
				1 = Always trigger the Write Sequencer on sample-rate detection			
	3:1	LRCLK_SRC[2:0]	000	Automatic Sample-Rate Detection source			
				000 = AIF1LRCLK			
				010 = AIF2LRCLK			
				100 = AIF3LRCLK			
				All other codes are reserved			
	0	RATE_EST_ENA	0	Automatic Sample-Rate Detection control			
				0 = Disabled			
				1 = Enabled			
R339 (0x0153)	4:0	SAMPLE_RATE_	0x00	Automatic Detection Sample Rate A			
Rate_Estimator_2		DETECT_A[4:0]		(Up to four different sample rates can be configured for automatic detection.)			
				Field coding is same as SAMPLE_RATE_ <i>n</i> .			
R340 (0x0154)	4:0	SAMPLE_RATE_	0x00	Automatic Detection Sample Rate B			
Rate_Estimator_3		DETECT_B[4:0]		(Up to four different sample rates can be configured for automatic detection.)			
				Field coding is same as SAMPLE_RATE_ <i>n</i> .			
R341 (0x0155)	4:0	SAMPLE_RATE_	0x00	Automatic Detection Sample Rate C			
Rate_Estimator_4		DETECT_C[4:0]		(Up to four different sample rates can be configured for automatic detection.)			
				Field coding is same as SAMPLE_RATE_ <i>n</i> .			
R342 (0x0156)	4:0	SAMPLE_RATE_	0x00	Automatic Detection Sample Rate D			
Rate_Estimator_5		DETECT_D[4:0]		(Up to four different sample rates can be configured for automatic detection.)			
				Field coding is same as SAMPLE_RATE_ <i>n</i> .			



Register Address	Bit	Label	Default	Description
R352 (0x0160)	15:13	ASYNC_CLK_	000	ASYNCCLK Frequency (Read only)
Clocking_debug_5		FREQ_STS[2:0]		000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				100 = 98.304 MHz (90.3168 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related sample rates only (i.e., ASYNC_SAMPLE_RATE_n = 01XXX).
	12:9	ASYNC_CLK_	0000	ASYNCCLK Source (Read only)
		SRC_STS[3:0]		0000 = MCLK1
				0001 = MCLK2
				0010 = MCLK3
				0100 = FLL1 × 2
				0101 = FLL2 × 2
				1000 = AIF1BCLK
				1001 = AIF2BCLK
				1010 = AIF3BCLK
				1111 = FLL1
				All other codes are reserved
	6:4	SYSCLK_FREQ_	000	SYSCLK Frequency (Read only)
		STS[2:0]		000 = 6.144 MHz (5.6448 MHz)
				001 = 12.288 MHz (11.2896 MHz)
				010 = 24.576 MHz (22.5792 MHz)
				011 = 49.152 MHz (45.1584 MHz)
				100 = 98.304 MHz (90.3168 MHz)
				All other codes are reserved
				The frequencies in brackets apply for 44.1 kHz–related sample rates only (i.e., SAMPLE_RATE_ <i>n</i> = 01XXX).
	3:0	SYSCLK_SRC_	0000	SYSCLK Source (Read only)
		STS[3:0]		0000 = MCLK1
				0001 = MCLK2
				0010 = MCLK3
				0100 = FLL1 × 2
				0101 = FLL2 × 2
				1000 = AIF1BCLK
				1001 = AIF2BCLK
				1010 = AIF3BCLK
				1111 = FLL1
				All other codes are reserved

In AIF Slave Modes, it is important to ensure that the applicable clock domain (SYSCLK or ASYNCCLK) is synchronized with the associated external LRCLK. This can be achieved by selecting an MCLK*n* input that is derived from the same reference as the LRCLK, or can be achieved by selecting the external BCLK or LRCLK signal as a reference input to one of the FLLs, as a source for SYSCLK or ASYNCCLK.

If the AIF clock domain is not synchronized with the LRCLK, clicks arising from dropped or repeated audio samples occur, due to the inherent tolerances of multiple, asynchronous, system clocks. See Section 5.4 for further details on valid clocking configurations.

4.16.6 BCLK and LRCLK Control

The digital audio interfaces (AIF1–AIF3) use BCLK and LRCLK signals for synchronization. In Master Mode, these are output signals, generated by the CS42L92. In Slave Mode, these are input signals to the CS42L92. It is also possible to support mixed master/slave operation.



The BCLK and LRCLK signals are controlled as shown in Fig. 4-66. See Section 4.7 for details of the associated control fields.

Note that the BCLK and LRCLK signals are synchronized to SYSCLK or ASYNCCLK, depending upon the applicable clock domain for the respective interface. See Section 4.3.14 for further details.

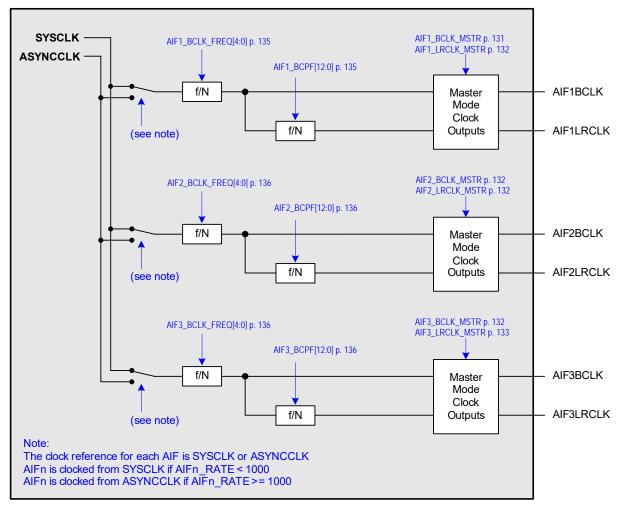


Figure 4-66. BCLK and LRCLK Control

4.16.7 Control Interface Clocking

Register map access is possible with or without a system clock—there is no requirement for SYSCLK, or any other system clock, to be enabled when accessing the register map. See Section 4.17 for details of control register access.

Timing specifications for each of the control interfaces are provided in Table 3-19–Table 3-21. In some applications, additional system-wide constraints must be observed to ensure control interface limits are not exceeded. These constraints need to be considered if any of the following conditions is true.

- SYSCLK is enabled and is < 22.5792 MHz
- Control-register access is scheduled at register address 0x8_0000 or above
- · Control-register access is scheduled on more than one of the control interfaces simultaneously

The control interface limits vary depending on the system clock (SYSCLK or DSPCLK) configuration, the address of the control register access, and on which control interfaces are being used.



Table 4-105 describes valid system conditions for accessing the codec registers (below 0x8_0000). The control interfaces must operate within the limits represented by one of the permitted configurations shown, in accordance with the applicable SYSCLK frequency.

Table 4-105	5. Maximum SPI/SLIMbus Clock	Speeds—Codec Register Access
-------------	------------------------------	------------------------------

SYSCLK Condition	SPI	SLIMbus	Description
SYSCLK is disabled, or SYSCLK \ge 22.5792 MHz	26 MHz	27 MHz	Full concurrent SPI/SLIMbus capability for codec register access.
SYSCLK = 12.288 MHz	26 MHz	—	SPI and SLIMbus operating in isolation.
	—	27 MHz	
	13 MHz	24.576 MHz	SPI and SLIMbus operating concurrently.
	12 MHz	27 MHz	
SYSCLK = 11.2896 MHz	26 MHz	—	SPI and SLIMbus operating in isolation.
	—	27 MHz	
	12 MHz	22.5792 MHz	SPI and SLIMbus operating concurrently.
	9 MHz	27 MHz	
SYSCLK < 11.2896 MHz	13 MHz	—	SPI and SLIMbus operating in isolation.
	—	27 MHz	

Notes:

• If SYSCLK < 11.2896 MHz, simultaneous register access via SPI/SLIMbus control interfaces should not be attempted.

• If SYSCLK is disabled, full concurrent SPI/SLIMbus capability for codec register access is supported.

- The SPI interface limits noted above are only applicable if the SPI interface is accessing codec registers. Options shown with "—" in the SPI column represent use cases where the SPI interface is either unused, or is being used to access the DSP registers.
- The SLIMbus interface limits noted above are only applicable if multibyte burst transfers of more than 8 bytes are scheduled. Options shown with "—" in the SLIMbus column represent use cases where SLIMbus is either unused, or is configured to support any combination of audio channels, and burst transfers ≤ 8 bytes.
- Register access via the I²C interface is supported at all times, regardless of the SLIMbus loading.

Table 4-106 describes valid system conditions for accessing the DSP firmware registers (0x8_0000 and above). The control interfaces must operate within the limits represented by one of the permitted configurations shown, in accordance with the applicable DSPCLK frequency.

Table 4-106.	Maximum SPI/SLIMbus	Clock Speeds—DSF	P Firmware Register Access

DSPCLK Condition	SPI	SLIMbus	Description
DSPCLK is disabled, or	26 MHz	—	One high speed interface.
DSPCLK ≥ 45 MHz	_	27 MHz	
	26 MHz	24.576 MHz	SPI and SLIMbus operating concurrently.
	21 MHz	27 MHz	
$24.576 \text{ MHz} \le \text{DSPCLK} < 45 \text{ MHz}$	26 MHz	—	SPI and SLIMbus operating in isolation.
	_	24.576 MHz	
12.288 MHz ≤ DSPCLK < 24.576 MHz	13 MHz	—	SPI and SLIMbus operating in isolation.
		12.288 MHz	

Notes:

If DSPCLK < 24.576MHz, simultaneous register access via SPI/SLIMbus control interfaces should not be attempted.

• If DSPCLK is disabled, the valid configurations are the same as for DSPCLK \ge 45MHz.

• The SPI interface limits noted above are only applicable if the SPI interface is accessing DSP registers. Options shown with "—" in the SPI column represent use cases where the SPI interface is either unused, or is being used to access the codec registers.

- The SLIMbus interface limits noted above are only applicable if multibyte burst transfers of more than 8 bytes are scheduled. Options shown with "—" in the SLIMbus column represent use cases where SLIMbus is either unused, or is configured to support any combination of audio channels, and burst transfers ≤ 8 bytes.
- Register access via the I²C interface is supported at all times, regardless of the SLIMbus loading.

4.16.8 Frequency-Locked Loop (FLL1, FLL2)

Two integrated FLLs are provided to support the clocking requirements of the CS42L92. These can be configured according to the available reference clocks and the application requirements. The reference clock may use a high frequency (e.g., 12.288 MHz) or low frequency (e.g., 32.768 kHz). The FLL is tolerant of jitter and may be used to generate a stable output clock from a less stable input reference.



4.16.8.1 Overview

The FLL characteristics are summarized in Table 3-11. In normal operation, the FLL output is frequency locked to an input clock reference. The FLL can be used to generate a free-running clock in the absence of any external reference, as described in Section 4.16.8.6.

4.16.8.2 FLL Enable

The FLL is enabled by setting FLLn_ENA (where n = 1 or 2 for the corresponding FLL). Note that the other FLL fields should be configured before enabling the FLL; the FLLn_ENA bit should be set as the final step of the FLLn enable sequence.

The FLL supports configurable free-running operation in FLL Hold Mode, using the FLL*n*_HOLD bit described in Section 4.16.8.6. If the FLL is enabled and FLL Hold Mode is selected, the configured output frequency is maintained without any input reference required. Note that, once the FLL output has been established, the FLL is always free running if the input reference clock is stopped, regardless of the FLL*n*_HOLD bit.

Note that, to disable the FLL while the input reference clock has stopped, FLL*n*_HOLD must be set before clearing FLL*n*_ENA.

When changing FLL settings, it is recommended to disable the FLL by clearing FLL*n*_ENA before updating the other register fields. It is possible to configure the FLL while the FLL is enabled, as described in Section 4.16.8.4. As a general rule, however, it is recommended to configure the FLL before setting FLL*n*_ENA.

The procedure for configuring the FLL is described in the following subsections. The description is applicable to FLL1 and FLL2; the associated control fields are described in Table 4-108 and Table 4-109 respectively.

4.16.8.3 Input Frequency Control

The main input reference is selected using FLL*n*_REFCLK_SRC. The available options are MCLK*n*, SLIMCLK, AIF*n*BCLK, or AIF*n*LRCLK.

The SLIMCLK reference is controlled by an adaptive divider on the external SLIMCLK input. The divider automatically adapts to the SLIMbus clock gear, to provide a constant reference frequency for the FLL—see Section 4.10.3.

The FLL*n*_REFCLK_DIV field controls a programmable divider on the input reference. The input can be divided by 1, 2, 4 or 8. The divider should be configured to bring each reference down to 13 MHz or below. For best performance, it is recommended that the highest possible frequency—within the 13 MHz limit—should be selected.

The FLL incorporates a reference-detection circuit for the main input clock. This ensures best FLL performance in the event of the main input clock being interrupted. If there is a possibility of the main input being interrupted while the FLL is enabled, then the reference-detection circuit must be enabled by setting FLLn_REFDET. The reference detection also provides input to the interrupt control circuit and can be used to trigger an interrupt event when the input reference is stopped—see Section 4.15.

4.16.8.4 Output Frequency Control

The FLL output frequency, F_{FLL}, relative to the main input reference F_{REF}, is a function of:

- The frequency ratio set by FLLn_FB_DIV
- The real number represented by N.K. (N = integer; K = fractional portion, i.e., < 1)

The output frequency must be in the range 45-50 MHz.

If the FLL is selected as SYSCLK or ASYNCCLK source, the respective F_{FLL} frequency must be exactly 49.152 MHz (for 48 kHz–related sample rates) or 45.1584 MHz (for 44.1 kHz–related sample rates).

- If FLL2 is selected as SYSCLK or ASYNCCLK source, the respective clock frequency is equal to F_{FLL} × 2.
- If FLL1 is selected as SYSCLK or ASYNCCLK source, two different frequencies are available. Typical use cases should select the higher frequency (F_{FLL} × 2); a lower frequency (F_{FLL}) is available to support low-power always-on use cases.



If the FLL is selected as DSPCLK source, the following frequency options are supported:

- If FLL2 is selected as DSPCLK source, the DSPCLK frequency is in the range 135–150 MHz. The frequency is equal to F_{FLL} × 3.
- If FLL1 is selected as DSPCLK source, two different frequencies are available. Typical use cases should select the higher frequency (F_{FLL} × 3); a lower frequency (F_{FLL}) is available to support low-power always-on use cases.
- Note that the DSPCLK can be divided to lower frequencies for clocking the DSP core.

The FLL clock can be configured as a GPIO output; a programmable divider supports division ratios in the range 2 through 127, enabling a wide range of GPIO clock output frequencies.

Note: The chosen F_{FLL} frequency can be used to support multiple outputs simultaneously (e.g., SYSCLK, DSPCLK, and GPIO).

To configure the FLL output frequency, it must be determined whether Integer Mode or Fractional Mode is required.

- If the ratio F_{FLL} / F_{REF} is an integer, then Integer Mode applies
- If the ratio F_{FLL} / F_{REF} is not an integer, then Fractional Mode applies

The input reference must be identified in one of three frequency ranges.

- If F_{REF} < 192 kHz, this is *low* clock frequency
- If $F_{REF} \ge 192$ kHz and FREF < 1.152 MHz, this is *mid* clock frequency
- If $F_{REF} \ge 1.152$ MHz, this is *high* clock frequency

Note: F_{REF} is the input frequency, after division by FLL*n*_REFCLK_DIV, where applicable.

The FLL oscillator frequency, F_{OSC}, is set according to the applicable mode and input reference frequency.

- If Fractional Mode is used and F_{REF} is high frequency, then $F_{OSC} = F_{FLL} \times 6$
- Otherwise, F_{OSC} = F_{FLL}

The FLL oscillator frequency, F_{OSC}, is set according to the following equation:

 $F_{OSC} = (F_{REF} \times N.K \times FLLn_FB_DIV)$

The FLLn_FB_DIV value should be configured according to the applicable mode and input reference frequency.

- If Integer Mode is used and F_{REF} is low frequency, then FLLn_FB_DIV should be set to 4
- If Fractional Mode is used and F_{REF} is low frequency, then FLLn_FB_DIV should be set to 256
- Otherwise, FLLn_FB_DIV should be set to 1

The value of N.K can be determined as follows:

 $N.K = F_{OSC} / (FLLn_FB_DIV \times F_{REF})$

The calculated value of N must lie within a valid range, according to the applicable mode.

- If Integer Mode is used, N is valid in the range 1-1023
- If Fractional Mode is used, N is valid in the range 4–255

If the calculated value of N is too high, a higher FLL*n*_FB_DIV is required. If the calculated value of N is too low, a lower FLL*n*_FB_DIV is required. It is recommended to adjust the FLL*n*_FB_DIV value by multiplying or dividing by 2 until a valid N is achieved.

The value of N is held in FLLn_N.

The value of K is determined by the ratio FLL*n*_THETA / FLL*n*_LAMBDA. In Fractional Mode, the FLL*n*_THETA and FLL*n*_LAMBDA fields can be derived as described in Section 4.16.8.5.

The FLL*n*_N, FLL*n*_THETA, and FLL*n*_LAMBDA fields are all coded as integers (LSB = 1).



When changing FLL settings, it is recommended to disable the FLL by clearing FLL*n*_ENA before updating the other register fields. If the FLL settings or input reference are changed without disabling the FLL, the FLL Hold Mode must be selected before writing to any other FLL control fields. FLL Hold Mode is selected by setting FLL*n*_HOLD.

If the FLL control fields are written while the FLL is enabled (FLLn_ENA = 1), the new values are only effective when a 1 is written to FLLn_CTRL_UPD. This makes it possible to update the FLL configuration fields simultaneously, without disabling the FLL.

To change FLL settings without disabling the FLL, the recommended control sequence is:

- Select FLL Hold Mode (FLL*n*_HOLD = 1)
- Write to the FLL control fields
- Update the FLL control registers (write 1 to FLL*n*_CTRL_UPD)
- Disable FLL Hold Mode (FLL*n*_HOLD = 0)

Note that, if the FLL is disabled, the FLL control fields can be updated without writing to FLL*n*_CTRL_UPD.

The FLL*n*_PD_GAIN_FINE, FLL*n*_PD_GAIN_COARSE, FLL*n*_FD_GAIN_FINE, FLL*n*_FD_GAIN_COARSE, FLL*n*_HP, and FLLn_CLK_VCO_FAST_SRC fields should be configured as described in Table 4-107.

Note: When writing to the FLL*n*_CLK_VCO_FAST_SRC or FLL*n*_HP fields, take care not to change other nonzero bits that are configured at the same register address.

Condition	FLL <i>n_</i> PD_ GAIN_FINE	FLL <i>n</i> _PD_ GAIN_ COARSE	FLL <i>n</i> _FD_ GAIN_FINE	FLL <i>n_</i> FD_ GAIN_ COARSE	FLL <i>n_</i> LOCKDET_ THR	FLL <i>n_</i> CLK_ VCO_ FAST_SRC	FLL <i>n</i> _HP
Low clock frequency Mid clock frequency High clock frequency	0x2	0x3 0x2 0x1	0xF 0xF 0xF	0x0 0x2 0x0	0x2 0x8 0x8	_	_
Integer Mode Fractional Mode, Low clock frequency Fractional Mode, Mid clock frequency Fractional Mode, High clock frequency		_	_	_	_	0x0 0x0 0x0 0x3	_
Integer Mode Fractional Mode			—		—	—	0x0 0x3

Table 4-107. FLLn Control Field Settings

4.16.8.5 Calculation of Theta and Lambda

In Fractional Mode, FLLn_THETA and FLLn_LAMBDA are calculated with the following steps:

1. Calculate GCD(FLL) using the Greatest Common Denominator function:

 $GCD(FLL) = GCD(FLLn_FB_DIV \times F_{REF}, F_{OSC}),$

where GCD(x, y) is the greatest common denominator of x and y.

F_{REF} is the input frequency, after division by FLL*n*_REFCLK_DIV, where applicable.

2. Calculate FLL*n*_THETA and FLL*n*_LAMBDA using the following equations:

 $\label{eq:FLLn_THETA} = (F_{OSC} - (FLL_N \times FLLn_FB_DIV \times F_{REF})) / GCD(FLL) \\ FLLn_LAMBDA = (FLLn_FB_DIV \times F_{REF}) / GCD(FLL) \\ \end{array}$

Notes: The values of GCD(FLL), FLL*n*_THETA, and FLL*n*_LAMBDA should be calculated using the applicable frequency values in Hz (i.e., not kHz or MHz).

In Fractional Mode, the values of FLL*n*_THETA and FLL*n*_LAMBDA must be coprime (i.e., not divisible by any common integer). The calculation above ensures that the values are coprime.

The value of K must be less than 1 (i.e., FLL*n*_THETA must be less than FLL*n*_LAMBDA).



4.16.8.6 FLL Hold Mode

FLL Hold Mode enables the FLL to generate a clock signal even if no external reference is available, such as when the normal input reference has been interrupted during a standby or start-up period. FLL Hold Mode is selected by setting FLL*n*_HOLD.

If the FLL is enabled and FLL Hold Mode is selected, the normal feedback mechanism of the FLL is halted and the FLL oscillates independently of the external input references—the FLL output frequency remains unchanged if FLL Hold Mode is enabled.

If the FLL is enabled and the input reference clock is stopped, the loop always runs freely, regardless of the FLL*n*_HOLD setting. If FLL*n*_HOLD = 0, the FLL relocks to the input reference whenever it is available.

If the FLL configuration or input reference are changed without disabling the FLL, the FLL Hold Mode must be selected before writing to any other FLL control fields—see Section 4.16.8.4.

The free-running FLL clock may be selected as the SYSCLK, ASYNCCLK, or DSPCLK source, as shown in Fig. 4-65.

4.16.8.7 FLL Control Registers

. . .

The FLL1 control registers are described in Table 4-108.

Example settings for a variety of reference frequencies and output frequencies are shown in Section 4.16.8.10.

Register Address	Bit	Label	Default	Description				
R369 (0x0171)	3:0	FLL1_REFCLK_	0000	FLL1 Clock source				
FLL1_Control_1		SRC[3:0]		0000 = MCLK1	1000 = AIF1BCLK	1101 = AIF2LRCLK		
				0001 = MCLK2	1001 = AIF2BCLK	1110 = AIF3LRCLK		
				0010 = MCLK3	1010 = AIF3BCLK	All other codes are		
				0011 = SLIMCLK	1100 = AIF1LRCLK	reserved		
	2	FLL1_HOLD	1	FLL1 Hold Mode Enabl	e			
				0 = Disabled				
				1 = Enabled				
				setting is maintained.	hanism is halted in FLL Hold	Mode, and the latest integrator		
	0	FLL1_ENA	0	FLL1 Enable				
				0 = Disabled				
				1 = Enabled				
					he final step of the FLL1 enab	le sequence.		
R370 (0x0172)	15	FLL1_CTRL_UPD	0	FLL1 Control Update				
FLL1_Control_2					•	. (Only valid if FLL1_ENA = 1)		
	9:0	FLL1_N[9:0]	0x004	FLL1 Integer multiply for	or F _{REF}			
				Coded as LSB = 1.				
R371 (0x0173)	15:0	FLL1_	0x0000	FLL1 Fractional multipl				
FLL1_Control_3		THETA[15:0]		•	ultiply) part of the FLL1_THET	A/FLL1_LAMBDA ratio.		
				Coded as LSB = 1.				
R372 (0x0174)	15:0	FLL1_	0x0000	FLL1 Fractional multipl				
FLL1_Control_4		LAMBDA[15:0]		•	dividing) part of the FLL1_TH	ETA/FLL1_LAMBDA ratio.		
				Coded as LSB = 1.	-			
R373 (0x0175)	9:0	FLL1_FB_DIV[9:0]	0x0001	FLL1 Clock Feedback	ratio			
FLL1_Control_5				Coded as LSB = 1.	· · · ·			
R374 (0x0176)	15	FLL1_REFDET	1	FLL1 Reference Detect	t control			
FLL1_Control_6				0 = Disabled				
				1 = Enabled				
	7:6	FLL1_REFCLK_	00	FLL1 Clock Reference				
		DIV[1:0]		00 = 1	10 = 4			
				01 = 2	11 = 8			
				MCLK (or other input re	eference) must be divided dov	vn to \leq 13 MHz.		

Table 4-108. FLL1 Register Map



Register Address	Bit	Label	Default		Description			
R376 (0x0178)	15:12	FLL1_PD_GAIN_	0x2	FLL1 Phase Detector Gain 2				
FLL1_Control_8		FINE[3:0]		Gain is 2 ^{-X} , where X is FLL1_PD_GAIN_FINE in 2's complement coding.				
				0000 = 1	0110 = 2 ⁻⁶	1100 = 16		
				0001 = 0.5	0111 = 2 ⁻⁷	1101 = 8		
				0010 = 0.25	1000 = 256	1110 = 4		
				0011 = 0.125	1001 = 128	1111 = 2		
				0100 = 2-4	1010 = 64			
				0101 = 2 ⁻⁵	1011 = 32			
	11:8	FLL1_PD_GAIN_	0x1	FLL1 Phase Detector	Gain 1			
		COARSE[3:0]		Gain is 2 ^{_X} , where X is	s FLL1_PD_GAIN_COARSI	E in 2's complement coding.		
				0000 = 1	0110 = 2-6	1100 = 16		
				0001 = 0.5	0111 = 2 ⁻⁷	1101 = 8		
				0010 = 0.25	1000 = 256	1110 = 4		
				0011 = 0.125	1001 = 128	1111 = 2		
				0100 = 2-4	1010 = 64			
				0101 = 2 ⁻⁵	1011 = 32			
	7:4	FLL1 FD GAIN	0xF	FLL1 Frequency Deter	ctor Gain 2			
		FINE[3:0]		Gain is 2 ^{-X} , where X is	s FLL1_FD_GAIN_FINE in i	nteger coding.		
				0000 = 1	0011 = 0.125	$1110 = 2^{-14}$		
				0001 = 0.5		1111 = Reserved		
				0010 = 0.25	1101 = 2 ⁻¹³			
	3:0	FLL1_FD_GAIN_	0x0	FLL1 Frequency Dete	ctor Gain 1			
		COARSE[3:0]		Gain is 2 ^{_X} , where X is	s FLL1_FD_GAIN_COARSE	E in 2's complement coding.		
				0000 = 1	0110 = 2 ⁻⁶	1100 = 16		
				0001 = 0.5	0111 = 2 ⁻⁷	1101 = 8		
				0010 = 0.25	1000 = 256	1110 = 4		
				0011 = 0.125	1001 = 128	1111 = 2		
				0100 = 2-4	1010 = 64			
				0101 = 2 ⁻⁵	1011 = 32			
R378 (0x017A)	15:14	FLL1_HP[1:0]	00	FLL1 Fractional Mode	control			
FLL1_Control_10				00 = Integer mode	10 = Reserved			
				01 = Reserved	11 = Fractional Mode			
R379 (0x017B)	4:1	FLL1_LOCKDET_	0x8	FLL1 Lock Detect thre	shold			
FLL1_Control_11		THR[3:0]		Valid from 0x0 (low thr	eshold) to 0xF (high thresho	old)		
	0	FLL1_LOCKDET	1	FLL1 Lock Detect ena	bled			
				0 = Disabled				
				1 = Enabled				
R381 (0x017D)	1:0	FLL1_CLK_VCO_	0x0	FLL1 Oscillator Freque	ency Control			
FLL1_Digital_		FAST_SRC[1:0]		00 = 45–50 MHz	10 = Reserved			
Test_1				01 = Reserved	11 = 270–300 MHz			

Table 4-108. FLL1 Register Map (Cont.)



The FLL2 control registers are described in Table 4-109.

Table 4-109. FLL2 Register Map

Register Address	Bit	Label	Default		Description	
R401 (0x0191)	3:0	FLL2_REFCLK_	0111	FLL2 Clock source		
FLL2_Control_1		SRC[3:0]		0000 = MCLK1	1000 = AIF1BCLK	1101 = AIF2LRCLK
				0001 = MCLK2	1001 = AIF2BCLK	1110 = AIF3LRCLK
				0010 = MCLK3	1010 = AIF3BCLK	All other codes are
				0011 = SLIMCLK	1100 = AIF1LRCLK	reserved
	2	FLL2_HOLD	1	FLL2 Hold Mode Enable		
				0 = Disabled		
				1 = Enabled		
					anism is halted in FLL Hold	Mode, and the latest integrator
				setting is maintained.		
	0	FLL2_ENA	0	FLL2 Enable		
				0 = Disabled		
				1 = Enabled		
					final step of the FLL2 enab	le sequence.
R402 (0x0192)	15	FLL2_CTRL_UPD	0	FLL2 Control Update		
FLL2_Control_2						(Only valid if FLL2_ENA = 1)
	9:0	FLL2_N[9:0]	0x004	FLL2 Integer multiply for	F _{REF}	
				Coded as LSB = 1.		
R403 (0x0193)	15:0	FLL2_	0x0000	FLL2 Fractional multiply 1		
FLL2_Control_3		THETA[15:0]			iply) part of the FLL2_THET	A/FLL2_LAMBDA ratio.
				Coded as LSB = 1.		
R404 (0x0194)	15:0	FLL2_	0x0000	FLL2 Fractional multiply 1		
FLL2_Control_4		LAMBDA[15:0]			viding) part of the FLL2_TH	ETA/FLL2_LAMBDA ratio.
				Coded as LSB = 1.		
R405 (0x0195)	9:0	FLL2_FB_DIV[9:0]	0x0001	FLL2 Clock Feedback rat	tio	
FLL2_Control_5				Coded as LSB = 1.		
R406 (0x0196)	15	FLL2_REFDET	1	FLL2 Reference Detect c	control	
FLL2_Control_6				0 = Disabled		
				1 = Enabled		
	7:6	FLL2_REFCLK_		FLL2 Clock Reference di		
		DIV[1:0]		00 = 1	10 = 4	
				01 = 2	11 = 8	
				MCLK (or other input refe	erence) must be divided dov	vn to \leq 13 MHz.



Register Address	Bit	Label	Default		Description	
R408 (0x0198)	15:12	FLL2_PD_GAIN_	0x2	FLL2 Phase Detector G	Gain 2	
FLL2_Control_8		FINE[3:0]		Gain is 2 ^{_X} , where X is	FLL2_PD_GAIN_FINE in 2	's complement coding.
				0000 = 1	0110 = 2 ⁻⁶	1100 = 16
				0001 = 0.5	0111 = 2-7	1101 = 8
				0010 = 0.25	1000 = 256	1110 = 4
				0011 = 0.125	1001 = 128	1111 = 2
				0100 = 2-4	1010 = 64	
				0101 = 2 ⁻⁵	1011 = 32	
	11:8	FLL2_PD_GAIN_	0x1	FLL2 Phase Detector G	Gain 1	
		COARSE[3:0]		Gain is 2 ^{_X} , where X is	FLL2_PD_GAIN_COARSE	in 2's complement coding.
				0000 = 1	0110 = 2-6	1100 = 16
				0001 = 0.5	0111 = 2 ⁻⁷	1101 = 8
				0010 = 0.25	1000 = 256	1110 = 4
				0011 = 0.125	1001 = 128	1111 = 2
				0100 = 2-4	1010 = 64	
				0101 = 2 ⁻⁵	1011 = 32	
	7:4	FLL2 FD GAIN				
		FINE[3:0]		FLL2 Frequency Detec Gain is 2 ^{_X} , where X is	FLL2 FD GAIN FINE in in	nteger coding.
				0000 = 1	0011 = 0.125	1110 = 2-14
				0001 = 0.5		1111 = Reserved
				0010 = 0.25	1101 = 2 ⁻¹³	
	3:0	FLL2_FD_GAIN_	0x0	FLL2 Frequency Detec	tor Gain 1	
		COARSE[3:0]		Gain is 2 ^{_X} , where X is	FLL2 FD GAIN COARSE	in 2's complement coding.
				0000 = 1	0110 = 2 ⁻⁶	1100 = 16
				0001 = 0.5	0111 = 2 ⁻⁷	1101 = 8
				0010 = 0.25	1000 = 256	1110 = 4
				0011 = 0.125	1001 = 128	1111 = 2
				0100 = 2-4	1010 = 64	
				0101 = 2 ⁻⁵	1011 = 32	
R410 (0x019A)	15:14	FLL2_HP[1:0]	00	FLL2 Fractional Mode	control	
FLL2 Control 10				00 = Integer mode	10 = Reserved	
				01 = Reserved	11 = Fractional Mode	
R411 (0x019B)	4:1	FLL2_LOCKDET_	0x8	FLL2 Lock Detect thres	shold	
FLL2_Control_11		THR[3:0]		Valid from 0x0 (low thre	eshold) to 0xF (high thresho	ld)
	0	FLL2_LOCKDET	1	FLL2 Lock Detect enab	/ / /	
		_		0 = Disabled		
				1 = Enabled		
R413 (0x019D)	1:0	FLL2_CLK_VCO_	0x0	FLL2 Oscillator Freque	ncy Control	
FLL2_Digital_		FAST_SRC[1:0]		00 = 45–50 MHz	10 = Reserved	
Test_1				01 = Reserved	11 = 270–300 MHz	

Table 4-109. FLL2 Register Map (Cont.)

4.16.8.8 FLL Interrupts and GPIO Output

For each FLL, the CS42L92 provides status signals that indicate whether the input reference is present and whether FLL lock has been achieved (i.e., the FLL is locked to the input reference signal).

To enable the FLL lock indication, the FLL*n*_LOCKDET bit must be set. The FLL lock condition is measured with respect to a configurable threshold that is set using FLL*n*_LOCKDET_THR. Note that the FLL*n*_LOCKDET_THR field controls the lock indication only—it does not control the behavior of the FLL.

To enable the FLL input reference indication, the FLL*n*_REFDET bit must be set.

The FLL status signals are inputs to the interrupt control circuit and can be used to trigger an interrupt event when the input reference is stopped or when the FLL lock status changes. See Section 4.15. Note that these interrupt signals are debounced and require clocking to be present in order to assert the respective interrupt—either the 32-kHz clock or the SYSCLK signal must be enabled to trigger an interrupt from the FLL signals.



The FLL lock signal can be output directly on a GPIO pin as an external indication of the FLL status. See Section 4.14 to configure a GPIO pin for these functions. (These GPIO outputs are not debounced and do not require clocking to be present.)

Clock output signals derived from the FLL can be output on a GPIO pin. See Section 4.14 to configure a GPIO pin for this function.

4.16.8.9 Example FLL Calculation

The following example illustrates how to derive the FLL1 register fields to generate an FLL output frequency (F_{FLL}) of 49.152 MHz from a 12.000-MHz reference clock (F_{REF}). This is suitable for generating SYSCLK at 98.304 MHz and DSPCLK at 147.456 MHz.

- 1. Set FLL1_REFCLK_DIV to generate $F_{REF} \le 13$ MHz: FLL1_REFCLK_DIV = 00 (divide by 1)
- 2. Determine if Integer Mode or Fractional Mode is required:
 - $\mathsf{F}_{\mathsf{FLL}}$ / $\mathsf{F}_{\mathsf{REF}}$ is 4.096. Therefore, Fractional Mode applies.
- 3. Identify the input clock frequency range:
 - $F_{REF} \ge 1.152$ MHz. This is *high* clock frequency.
- 4. Calculate the FLL oscillator frequency, FOSC:

In Fractional Mode, with high clock frequency input, $F_{OSC} = F_{FLL} \times 6 = 294.912 \text{ MHz}$

- 5. Select the required value of FLL1_FB_DIV:
 - In Fractional Mode, with high clock frequency input, FLL1_FB_DIV = 1
- Calculate N.K as given by N.K = F_{OSC} / (FLL1_FB_DIV × F_{REF}): N.K = 294912000 / (1 × 12000000) = 24.576
- 7. Confirm that the calculated value of N is within the valid range for fractional mode (4-255).
- 8. Determine FLL1_N from the integer portion of N.K:
 - FLL1_N = 24 (0x018)
- Determine GCD(FLL), as given by GCD(FLL) = GCD(FLL1_FB_DIV × F_{REF}, F_{OSC}): GCD(FLL) = GCD(1 × 1200000, 294912000) = 96000
- 10.Determine FLL1_THETA, as given by FLL1_THETA = (F_{OSC} (FLL1_N × FLL1_FB_DIV × F_{REF})) / GCD(FLL): FLL1_THETA = (294912000 – (24 × 1 × 12000000)) / 96000 FLL1_THETA = 72 (0x0048)
- 11. Determine FLL1_LAMBDA, as given by FLL1_LAMBDA = (FLL1_FB_DIV x F_{REF}) / GCD(FLL): FLL1_LAMBDA = (1 × 1200000) / 96000 FLL1_LAMBDA = 125 (0x007D)
- 12. Determine other FLL settings as specified in Table 4-107 for Fractional Mode and high clock frequency input:

FLL1_PD_GAIN_FINE = 0x2 FLL1_PD_GAIN_COARSE = 0x1 FLL1_FD_GAIN_FINE = 0xF FLL1_FD_GAIN_COARSE = 0x0 FLL1_CLK_VCO_FAST_SRC = 0x3 FLL1_HP = 0x3



4.16.8.10 Example FLL Settings

Table 4-110 shows FLL settings for generating an output frequency (F_{FLL}) of 49.152 MHz from a variety of low- and high-frequency reference inputs. This is suitable for generating SYSCLK at 98.304 MHz and DSPCLK at 147.456 MHz.

F _{SOURCE}	F _{FLL} (MHz)	F _{REF} Divider ¹	FB_DIV ¹	N.K ²	FLL <i>n</i> _N	FLL <i>n_</i> THETA	FLL <i>n_</i> LAMBDA
32.000 kHz	49.152	1	4	384	0x180	0x0000	0x0001
32.768 kHz	49.152	1	4	375	0x177	0x0000	0x0001
44.100 kHz	49.152	1	256	4.3537415	0x004	0x0034	0x0093
48 kHz	49.152	1	4	256	0x100	0x0000	0x0001
128 kHz	49.152	1	4	96	0x060	0x0000	0x0001
9.6 MHz	49.152	1	1	30.72	0x01E	0x0012	0x0019
10 MHz	49.152	1	1	29.4912	0x01D	0x0133	0x0271
11.2896 MHz	49.152	1	1	26.12245	0x01A	0x0006	0x0031
12.000 MHz	49.152	1	1	24.576	0x018	0x0048	0x007D
12.288 MHz	49.152	1	1	4	0x004	0x0000	0x0001
13.000 MHz	49.152	1	1	22.68554	0x016	0x045A	0x0659
19.200 MHz	49.152	2	1	30.72	0x01E	0x0012	0x0019
22.5792 MHz	49.152	2	1	26.12245	0x01A	0x0006	0x0031
24 MHz	49.152	2	1	24.576	0x018	0x0048	0x007D
24.576 MHz	49.152	2	1	4	0x004	0x0000	0x0001
26 MHz	49.152	2	1	22.68554	0x016	0x045A	0x0659

Table 4-110.	Example	FLL	Settinas

1.See Table 4-108 and Table 4-109 for the coding of the FLL*n*_REFCLK_DIV and FLL*n*_FB_DIV fields. 2.N.K values are represented in the FLL*n*_N, FLL*n*_THETA, and FLL*n*_LAMBDA fields.

4.17 Control Interface

The CS42L92 is controlled by read/write access to its control registers. The control interface supports 2-wire (I²C) and 4-wire (SPI) modes. Note that the SLIMbus interface also supports read/write access to the CS42L92 control registers; see Section 4.10.

The CS42L92 executes a boot sequence following power-on reset, hardware reset, software reset, or wake-up from Sleep Mode. Note that control register writes should not be attempted until the boot sequence has completed. See Section 4.22 for further details.

The control interface function can be supported with or without system clocking—there is no requirement for SYSCLK, or any other system clock, to be enabled when accessing the register map.

Timing specifications for each of the control interfaces are provided in Table 3-19–Table 3-21. In some applications, additional system-wide constraints must be observed to ensure control interface limits are not exceeded. Full details of these requirements are provided in Section 4.16.7. These constraints need to be considered if any of the following conditions is true.

- SYSCLK is enabled and is < 22.5792 MHz
- Control-register access is scheduled at register address 0x80000 or above
- · Control-register access is scheduled on more than one of the control interfaces simultaneously

The control interface can be configured as a 2-wire (I²C) or 4-wire (SPI) interface. The mode is determined by the logic level on the CIFMODE pin, as described in Table 4-111.



CIFMODE	Interface Mode	Pin Functions
Logic 1	Four-wire (SPI) interface	CIFMISO—Data output
		CIFMOSI—Data input
		CIFSCLK—Interface clock input
		CIFSS—Slave select input
Logic 0	Two-wire (I ² C) interface	CIFSCLK—Interface clock input
		CIFSDA—Data input/output

Table 4-111. CS42L92 Control Interface Summary

Note: The CIFMOSI and CIFSDA functions are multiplexed on a dual-function pin.

An integrated pull-down resistor is provided on the CIF1MISO pin. This provides a flexible capability for interfacing with other devices. The pull-down is configured using the CIF1MISO_PD bit, as described in Table 4-112.

Table 4-112. Control Interface Pull-Down

Register Address	Bit	Label	Default	Description
R8 (0x0008)	7	CIF1MISO_PD	0	CIFMISO Pull-Down Control
Ctrl_IF_CFG_1				0 = Disabled
				1 = Enabled

A detailed description of the I²C and SPI interface modes is provided in the following sections.

4.17.1 Four-Wire (SPI) Control Mode

The SPI control interface mode is supported using the CIFSS, CIFSCLK, CIFMOSI, and CIFMISO pins.

In write operations (R/W = 0), the MOSI pin input is driven by the controlling device.

In read operations (R/W = 1), the MOSI pin is ignored following receipt of the valid register address.

If \overline{SS} is asserted (Logic 0), the MISO output is actively driven when outputting data and is high impedance at other times. If \overline{SS} is not asserted, the MISO output is high impedance.

The high-impedance state of the MISO output allows the pin to be shared with other slaves. An internal pull-down resistor can be enabled on the MISO pin, as described in Table 4-112.

Data transfers on the SPI interface must use the applicable message format, according to the register address space that is being accessed:

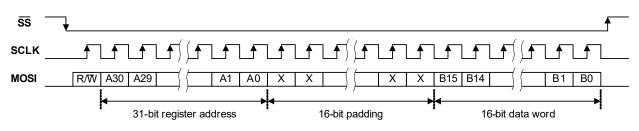
- When accessing register addresses below R12288 (0x3000), the applicable SPI protocol comprises a 31-bit register address and 16-bit data words.
- When accessing register addresses from R12888 (0x3000) upwards, the applicable SPI protocol comprises a 31-bit register address and 32-bit data words.
- Note that, in all cases, the complete SPI message protocol also includes a read/write bit and a 16-bit padding phase (see Fig. 4-67 and Fig. 4-68 below).

Continuous read and write modes enable multiple register operations to be scheduled faster than is possible with single register operations. In these modes, the CS42L92 automatically increments the register address at the end of each data word, for as long as SS is held low and SCLK is toggled. Successive data words can be input/output every 16 (or 32) clock cycles (depending on the applicable register address space).

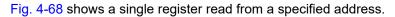
The SPI protocol is shown in Fig. 4-67 and Fig. 4-68. Note that 16-bit data words are shown, but the equivalent protocol also applies to 32-bit data words.

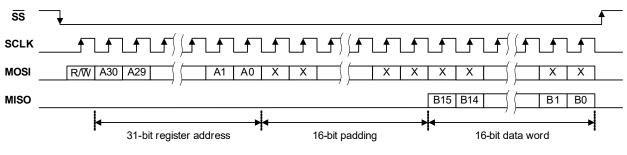


Fig. 4-67 shows a single register write to a specified address.











4.17.2 Two-Wire (I²C) Control Mode

The I²C control interface mode is supported using the CIFSCLK and CIFSDA pins.

In I²C Mode, the CS42L92 is a slave device on the control interface; SCLK is a clock input, while SDA is a bidirectional data pin. To allow arbitration of multiple slaves (and/or multiple masters) on the same interface, the CS42L92 transmits Logic 1 by tristating the SDA pin, rather than pulling it high. An external pull-up resistor is required to pull the SDA line high so that the Logic 1 can be recognized by the master.

In order to allow many devices to share a single two-wire control bus, every device on the bus has a unique 8-bit device ID (this is not the same as the address of each register in the CS42L92).

The CS42L92 device ID is 0011_0100 (0x34). Note that the LSB of the device ID is the read/write bit; this bit is set to Logic 1 for read and Logic 0 for write.

The CS42L92 operates as a slave device only. The controller indicates the start of data transfer with a high-to-low transition on SDA while SCLK remains high. This indicates that a device ID and subsequent address/data bytes follow. The CS42L92 responds to the start condition and shifts in the next 8 bits on SDA (8-bit device ID, including read/write bit, MSB first). If the device ID received matches the device ID of the CS42L92, the CS42L92 responds by pulling SDA low on the next clock pulse (ACK). If the device ID is not recognized or the R/W bit is set incorrectly, the CS42L92 returns to the idle condition and waits for a new start condition.

If the device ID matches the device ID of the CS42L92, the data transfer continues. The controller indicates the end of data transfer with a low-to-high transition on SDA while SCLK remains high. After receiving a complete address and data sequence the CS42L92 returns to the idle state and waits for another start condition. If a start or stop condition is detected out of sequence at any point during data transfer (i.e., SDA changes while SCLK is high), the device returns to the idle condition.



Data transfers on the I²C interface must use the applicable message format, according to the register address space that is being accessed:

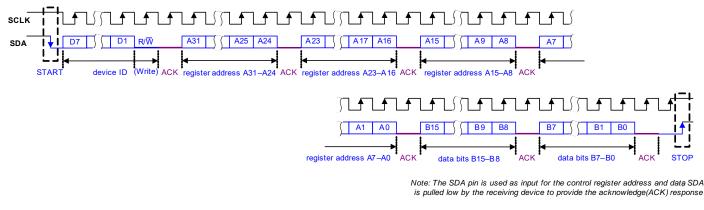
- When accessing register addresses below R12288 (0x3000), the applicable I²C protocol comprises a 32-bit register address and 16-bit data words.
- When accessing register addresses from R12888 (0x3000) upwards, the applicable I²C protocol comprises a 32-bit register address and 32-bit data words.
- Note that, in all cases, the complete I²C message protocol also includes a device ID, a read/write bit, and other signaling bits (see Fig. 4-69 and Fig. 4-70).

The CS42L92 supports the following read and write operations:

- Single write
- Single read
- Multiple write
- Multiple read

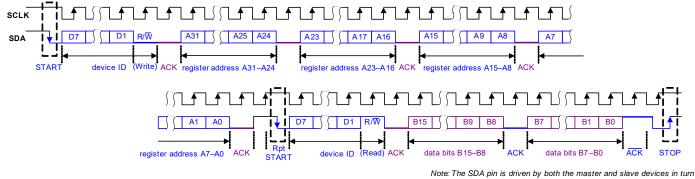
Continuous (multiple) read and write modes allow register operations to be scheduled faster than is possible with single register operations. In these modes, the CS42L92 automatically increments the register address after each data word. Successive data words can be input/output every 2 (or 4) data bytes, depending on the applicable register address space.

The I²C protocol for a single, 16-bit register write operation is shown in Fig. 4-69.





The I²C protocol for a single, 16-bit register read operation is shown in Fig. 4-70.



to transfer device address, register address, data and ACK responses

Figure 4-70. Control Interface I²C Register Read (16-Bit Data Words)

The control interface also supports other register operations; the interface protocol for these operations is shown in Fig. 4-71 through Fig. 4-74. The terminology used in the following figures is detailed in Table 4-113.



Note that 16-bit data words are shown in these illustrations. The equivalent protocol is also applicable to 32-bit words, with 4 data bytes transmitted (or received) instead of 2.

Terminology	Description
S	Start condition
Sr	Repeated start
А	Acknowledge (SDA low)
Ā	Not acknowledge (SDA high)
Р	Stop condition
R/W	Read/not write
	0 = Write; 1 = Read
[White field]	Data flow from bus master to CS42L92
[Gray field]	Data flow from CS42L92 to bus master

Table 4-113. Control Interface (I²C) Terminology

Fig. 4-71 shows a single register write to a specified address.

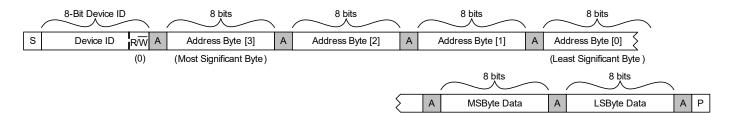


Figure 4-71. Single-Register Write to Specified Address

Fig. 4-72 shows a single register read from a specified address.

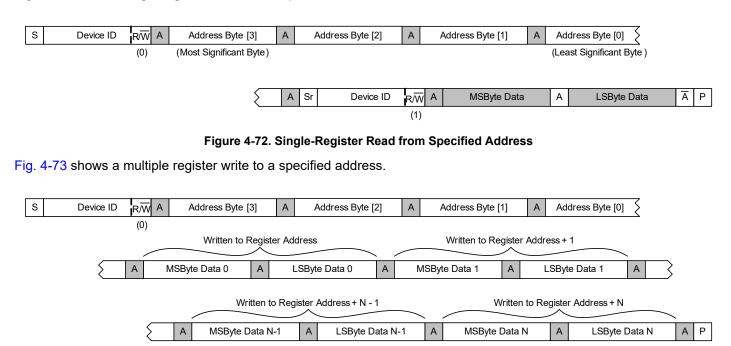


Figure 4-73. Multiple-Register Write to Specified Address



Fig. 4-74 shows a multiple register read from a specified address.

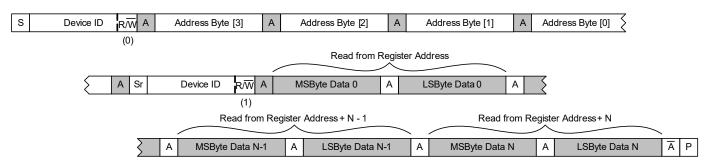


Figure 4-74. Multiple-Register Read from Specified Address

4.18 Control-Write Sequencer

The control-write sequencer is a programmable unit that forms part of the CS42L92 control interface logic. It provides the ability to perform a sequence of register-write operations with the minimum of demands on the host processor—the sequence may be initiated by a single operation from the host processor and then left to execute independently.

Default sequences for pop-suppressed start-up and shutdown of each headphone/earpiece output driver are provided (these are scheduled automatically when the respective output paths are enabled or disabled). Other control sequences can be programmed, and may be associated with sample-rate detection, DRC, MICDET clamp, or event logger status; these sequences are automatically scheduled whenever a corresponding event is detected.

When a sequence is initiated, the sequencer performs a series of predefined register writes. The start index of a control sequence within the sequencer's memory may be commanded directly by the host processor. The applicable start index for each of the sequences associated with sample-rate detection, DRC, MICDET clamp, or event logger status is held in a user-programmed control register.

The control-write sequencer may be triggered by a number of different events. Multiple sequences are queued if necessary, and each is scheduled in turn.

The control-write sequencer can be supported with or without system clocking—there is no requirement for SYSCLK or for any other system clock to be enabled when using the control-write sequencer. The timing accuracy of the sequencer operation is improved when SYSCLK is present, but the general functionality is supported with or without SYSCLK.

4.18.1 Initiating a Sequence

The fields associated with running the control-write sequencer are described in Table 4-114.

The CS42L92 provides 16 general-purpose trigger bits for the write sequencer to allow easy triggering of the associated control sequences. Writing 1 to the trigger bit initiates a control sequence, starting at the respective index position within the control-write sequencer memory.

The WSEQ_TRG1_INDEX field defines the sequencer start index corresponding to the WSEQ_TRG1 trigger control bit. Equivalent start index fields are provided for each trigger control bit, as described in Table 4-114. Note that a sequencer start index of 0x1FF causes the respective sequence to be aborted.

The general-purpose control sequences are undefined following power-on reset, a hardware reset, or a Sleep Mode transition. The general-purpose control sequences must be reconfigured by the host processor following any of these events. Note that all control sequences are maintained in the sequencer memory through software reset.

The write sequencer can also be commanded using control bits in register R22 (0x16). In this case, the write sequencer is enabled using the WSEQ_ENA bit and the index location of the first command in the sequence is held in the WSEQ_START_INDEX field. Writing 1 to the WSEQ_START bit commands the sequencer to execute a control sequence, starting at the specified index position. Note that, if the sequencer is already running, the WSEQ_START command is queued and executed when the sequencer becomes available.



Note: The mechanism for queuing multiple sequence requests has limitations when the WSEQ_START bit is used to trigger the write sequencer. If a sequence is initiated using the WSEQ_START bit, no other control sequences should be triggered until the sequence completes. The WSEQ_BUSY bit (described in Table 4-120) provides an indication of the sequencer status and can be used to confirm the sequence has completed.

Multiple control sequences triggered by any other method are queued if necessary, and scheduled in turn.

The write sequencer can be interrupted by writing 1 to the WSEQ_ABORT bit. Note that this command only aborts a sequence that is currently running; if other sequence commands are pending and not yet started, these sequences are not aborted by writing to the WSEQ_ABORT bit.

The write sequencer stores up to 508 register-write commands. These are defined in registers R12288 (0x3000) through R13302 (0x33F6). See Table 4-121 for a description of these registers.

Register Address	Bit	Label	Default	Description
R22 (0x0016)	11	WSEQ_ABORT	0	Writing 1 to this bit aborts the current sequence.
Write_Sequencer_ Ctrl_0	10	WSEQ_START	0	Writing 1 to this bit starts the write sequencer at the index location selected by WSEQ_START_INDEX. At the end of the sequence, this bit is reset by the write sequencer.
	9	WSEQ_ENA	0	Write Sequencer Enable
				0 = Disabled
				1 = Enabled
				Only applies to sequences triggered using the WSEQ_START bit.
	8:0	WSEQ_	0x000	Sequence Start Index. Contains the index location in the sequencer memory of the first
		START_		command in the selected sequence.
		INDEX[8:0]		Only applies to sequences triggered using the WSEQ_START bit.
				Valid from 0 to 507 (0x1FB).
R66 (0x0042)	15	WSEQ_TRG16	0	Write Sequence Trigger 16
Spare_Triggers	<u> </u>			Write 1 to trigger
	14	WSEQ_TRG15	0	Write Sequence Trigger 15
				Write 1 to trigger
	13	WSEQ_TRG14	0	Write Sequence Trigger 14
				Write 1 to trigger
	12	WSEQ_TRG13	0	Write Sequence Trigger 13
			-	Write 1 to trigger
	11	WSEQ_TRG12	0	Write Sequence Trigger 12
			-	Write 1 to trigger
	10	WSEQ_TRG11	0	Write Sequence Trigger 11
				Write 1 to trigger
	9	WSEQ_TRG10	0	Write Sequence Trigger 10
			-	Write 1 to trigger
	8	WSEQ_TRG9	0	Write Sequence Trigger 9
			-	Write 1 to trigger
	7	WSEQ_TRG8	0	Write Sequence Trigger 8
				Write 1 to trigger
	6	WSEQ_TRG7	0	Write Sequence Trigger 7
				Write 1 to trigger
	5	WSEQ_TRG6	0	Write Sequence Trigger 6
				Write 1 to trigger
	4	WSEQ_TRG5	0	Write Sequence Trigger 5
		14050 7504		Write 1 to trigger
	3	WSEQ_TRG4	0	Write Sequence Trigger 4
	0	WOED TROO	0	Write 1 to trigger
	2	WSEQ_TRG3	0	Write Sequence Trigger 3
	4			Write 1 to trigger
	1	WSEQ_TRG2	0	Write Sequence Trigger 2
	0			Write 1 to trigger
	0	WSEQ_TRG1	0	Write Sequence Trigger 1
				Write 1 to trigger

Table 4-114. Write Sequencer Control—Initiating a Sequence



Table 4-114.	Write Sec	uencer	Control-	_Initiating	a Sec	uence	(Cont.))
	Wille Deg	ucilicei	00111101-	-minuaung		lucince	00111.)	1

Register Address	Bit	Label	Default	Description
R75 (0x004B) Spare_Sequence_	8:0	WSEQ_TRG1_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG1 trigger.
Select_1 R76 (0x004C)	8:0	WSEQ_TRG2_	0x1FF	Valid from 0 to 507 (0x1FB). Write Sequence trigger 1 start index. Contains the index location in the sequencer
Spare_Sequence_ Select_2		INDEX[8:0]		memory of the first command in the sequence associated with the WSEQ_TRG2 trigger. Valid from 0 to 507 (0x1FB).
R77 (0x004D) Spare Sequence	8:0	WSEQ_TRG3_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG3 trigger.
Select_3				Valid from 0 to 507 (0x1FB).
R78 (0x004E) Spare_Sequence_ Select_4	8:0	WSEQ_TRG4_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG4 trigger. Valid from 0 to 507 (0x1FB).
R79 (0x004F) Spare_Sequence_ Select 5	8:0	WSEQ_TRG5_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG5 trigger. Valid from 0 to 507 (0x1FB).
R80 (0x0050) Spare_Sequence_ Select 6	8:0	WSEQ_TRG6_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG6 trigger. Valid from 0 to 507 (0x1FB).
 R89 (0x0059) Spare_Sequence_	8:0	WSEQ_TRG7_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG7 trigger.
Select_7 R90 (0x005A)	8:0	WSEQ_TRG8_	0x1FF	Valid from 0 to 507 (0x1FB). Write Sequence trigger 1 start index. Contains the index location in the sequencer
Spare_Sequence_ Select_8		INDEX[8:0]		memory of the first command in the sequence associated with the WSEQ_TRG8 trigger. Valid from 0 to 507 (0x1FB).
R91 (0x005B) Spare_Sequence_	8:0	WSEQ_TRG9_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG9 trigger.
Select_9 R92 (0x005C) Spare_Sequence_ Select_10	8:0	WSEQ_ TRG10_ INDEX[8:0]	0x1FF	Valid from 0 to 507 (0x1FB). Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG10 trigger. Valid from 0 to 507 (0x1FB).
R93 (0x005D) Spare_Sequence_ Select_11	8:0	WSEQ_ TRG11_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG11 trigger. Valid from 0 to 507 (0x1FB).
R94 (0x005E) Spare_Sequence_ Select_12	8:0	WSEQ_ TRG12_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG12 trigger. Valid from 0 to 507 (0x1FB).
R104 (0x0068) Spare_Sequence_ Select_13	8:0	WSEQ_ TRG13_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG13 trigger. Valid from 0 to 507 (0x1FB).
R105 (0x0069) Spare_Sequence_ Select_14	8:0	WSEQ_ TRG14_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG14 trigger. Valid from 0 to 507 (0x1FB).
R106 (0x006A) Spare_Sequence_ Select_15	8:0	WSEQ_ TRG15_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG15 trigger. Valid from 0 to 507 (0x1FB).
R107 (0x006B) Spare_Sequence_ Select_16	8:0	WSEQ_ TRG16_ INDEX[8:0]	0x1FF	Write Sequence trigger 1 start index. Contains the index location in the sequencer memory of the first command in the sequence associated with the WSEQ_TRG16 trigger. Valid from 0 to 507 (0x1FB).

4.18.2 Automatic Sample-Rate Detection Sequences

The CS42L92 supports automatic sample-rate detection on the digital audio interfaces (AIF1–AIF3) when operating in AIF Slave Mode. Automatic sample-rate detection is enabled by setting RATE_EST_ENA—see Table 4-104.



As many as four audio sample rates can be configured for automatic detection; these sample rates are selected using the SAMPLE_RATE_DETECT_*n* fields. If a selected audio sample rate is detected, the control-write sequencer is triggered. The applicable start index location within the sequencer memory is separately configurable for each detected sample rate.

The WSEQ_SAMPLE_RATE_DETECT_A_INDEX field defines the sequencer start index corresponding to the SAMPLE_ RATE_DETECT_A sample rate. Equivalent start index fields are defined for the other sample rates, as described in Table 4-115.

Note that a sequencer start index of 0x1FF causes the respective sequence to be aborted.

The automatic sample-rate detection control sequences are undefined following power-on reset, a hardware reset, or a Sleep Mode transition. The automatic sample-rate detection control sequences must be reconfigured by the host processor following any of these events. Note that all control sequences are maintained in the sequencer memory through software reset.

See Section 4.16 for further details of the automatic sample-rate detection function.

Register Address	Bit	Label	Default	Description
R97 (0x0061) Sample_Rate_ Sequence_Select_1	8:0	WSEQ_SAMPLE_ RATE_DETECT_ A_INDEX[8:0]	0x1FF	Sample Rate A Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with Sample Rate A detection. Valid from 0 to 507 (0x1FB).
R98 (0x0062) Sample_Rate_ Sequence_Select_2	8:0	WSEQ_SAMPLE_ RATE_DETECT_ B_INDEX[8:0]	0x1FF	Sample Rate B Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with Sample Rate B detection. Valid from 0 to 507 (0x1FB).
R99 (0x0063) Sample_Rate_ Sequence_Select_3	8:0	WSEQ_SAMPLE_ RATE_DETECT_ C_INDEX[8:0]	0x1FF	Sample Rate C Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with Sample Rate C detection. Valid from 0 to 507 (0x1FB).
R100 (0x0064) Sample_Rate_ Sequence_Select_4	8:0	WSEQ_SAMPLE_ RATE_DETECT_ D_INDEX[8:0]	0x1FF	Sample Rate D Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with Sample Rate D detection. Valid from 0 to 507 (0x1FB).

Table 4-115. Write Sequence Control—Automatic Sample-Rate Detection

4.18.3 DRC Signal-Detect Sequences

The DRC function within the CS42L92 digital core provides a configurable signal-detect function. This allows the signal level at the DRC input to be monitored and used to trigger other events.

The DRC signal-detect functions are enabled and configured using the fields described in Table 4-17 and Table 4-18 for DRC1 and DRC2 respectively.

A control-write sequence can be associated with a rising edge and/or a falling edge of the DRC1 signal-detect output. This is enabled by setting DRC1_WSEQ_SIG_DET_ENA, as described in Table 4-17.

Note that signal detection is supported on DRC1 and DRC2, but the triggering of the control-write sequencer is available on DRC1 only.

When the DRC signal-detect sequence is enabled, the control-write sequencer is triggered whenever the DRC1 signal-detect output transitions (high or low). The applicable start index location within the sequencer memory is separately configurable for each logic condition.

The WSEQ_DRC1_SIG_DET_RISE_SEQ_INDEX field defines the sequencer start index corresponding to a DRC1 signal-detect rising edge event, as described in Table 4-116. The WSEQ_DRC1_SIG_DET_FALL_SEQ_INDEX field defines the sequencer start index corresponding to a DRC1 signal-detect falling edge event.

Note that a sequencer start index of 0x1FF causes the respective sequence to be aborted.

The DRC signal-detect sequences cannot be independently enabled for rising and falling edges. Instead, a start index of 0x1FF can be used to disable the sequence for either edge, if required.



The DRC signal-detect control sequences are undefined following power-on reset, a hardware reset, or a Sleep Mode transition. The DRC signal-detect control sequences must be reconfigured by the host processor following any of these events. Note that all control sequences are maintained in the sequencer memory through software reset.

See Section 4.3.5 for further details of the DRC function.

Register Address	Bit	Label	Default	Description
R110 (0x006E) Trigger_ Sequence_ Select_32		WSEQ_DRC1_ SIG_DET_RISE_ INDEX[8:0]		DRC1 Signal-Detect (Rising) Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with DRC1 Signal-Detect (Rising) detection. Valid from 0 to 507 (0x1FB).
R111 (0x006F) Trigger_ Sequence_ Select_33		WSEQ_DRC1_ SIG_DET_FALL_ INDEX[8:0]		DRC1 Signal-Detect (Falling) Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with DRC1 Signal-Detect (Falling) detection. Valid from 0 to 507 (0x1FB).

Table 4-116. Write Sequencer Control—DRC Signal-Detect

4.18.4 MICDET Clamp Sequences

The CS42L92 supports external accessory detection functions, including the MICDET clamp circuits. The MICDET clamp status can be used to trigger the control-write sequencer. The MICDET clamps can be controlled using selectable logic conditions in respect of the JD*n* signals, as described in Table 4-85.

A control-write sequence can be associated with a rising edge and/or a falling edge of the MICDET clamp status. This is configured using the WSEQ_ENA_MICD_CLAMP_RISE and WSEQ_ENA_MICD_CLAMP_FALL bits, as described in Table 4-85.

If one of the selected JD*n* logic conditions is detected, the control-write sequencer is triggered. Note that these control-sequencer events are only valid if the clamp status changed in response to the JD*n* signals. The applicable start index location within the sequencer memory is separately configurable for the rising and falling edge conditions.

Note that, due to control logic that is shared between the two MICDET clamps, the option to control both clamps in response to the JD*n* signals cannot be supported at the same time. It is assumed that a maximum of one clamp is active at any time. Accordingly, only the active clamp is capable of triggering the control-write sequencer.

The WSEQ_MICD_CLAMP_RISE_INDEX field defines the sequencer start index corresponding to a MICDET clamp rising edge (clamp active) event, as described in Table 4-117. The WSEQ_MICD_CLAMP_FALL_INDEX field defines the sequencer start index corresponding to a MICDET clamp falling edge event.

Note that a sequencer start index of 0x1FF causes the respective sequence to be aborted.

The MICDET clamp control sequences are undefined following power-on reset, a hardware reset, or a Sleep Mode transition. The MICDET clamp control sequences must be reconfigured by the host processor following any of these events. Note that all control sequences are maintained in the sequencer memory through software reset.

See Section 4.12 for further details of the MICDET clamp status signals.

Register Address	Bit	Label	Default	Description
R102 (0x0066) Always_On_Triggers_ Sequence_Select_1		WSEQ_MICD_ CLAMP_RISE_ INDEX[8:0]		MICDET Clamp (Rising) Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with MICDET clamp (Rising) detection. Valid from 0 to 507 (0x1FB).
R103 (0x0067) Always_On_Triggers_ Sequence_Select_2		WSEQ_MICD_ CLAMP_FALL_ INDEX[8:0]		MICDET Clamp (Falling) Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with MICDET clamp (Falling) detection. Valid from 0 to 507 (0x1FB).

Table 4-117. Write Sequencer Control-MICDET Clamp

4.18.5 Event Logger Sequences

The CS42L92 provides an event log function, for monitoring and recording internal or external signals. The logged events are held in a FIFO buffer, from which the application software can read details of the detected logic transitions.



The control-write sequencer is automatically triggered whenever the NOT_EMPTY status of the event log buffer is asserted.

The WSEQ_EVENTLOG1_INDEX field defines the sequencer start index corresponding to the event logger, as described in Table 4-118.

Note that a sequencer start index of 0x1FF causes the respective sequence to be aborted.

The event logger control sequence is undefined following power-on reset, a hardware reset, or a Sleep Mode transition. The event logger control sequence must be reconfigured by the host processor following any of these events. Note that all control sequences are maintained in the sequencer memory through software reset.

See Section 4.5.1 for further details of the event logger.

Table 4-118.	Write Sequencer Control—Event Logger
14010 1 1101	

Register Address	Bit	Label	Default	Description
R120 (0x0078) Eventlog_ Sequence_ Select_1		WSEQ_ EVENTLOG1_ INDEX[8:0]		Event Log 1 Write Sequence start index. Contains the index location in the sequencer memory of the first command in the sequence associated with Event Log 1 FIFO Not-Empty detection. Valid from 0 to 507 (0x1FB).

4.18.6 Boot Sequence

The CS42L92 executes a boot sequence following power-on reset, hardware reset, software reset, or wake-up from Sleep Mode. The boot sequence configures the CS42L92 with factory-set trim (calibration) data. See Section 4.22 and Section 4.23 for further details.

The start index location of the boot sequence is 384 (0x180). See Table 4-123 for details of the write sequencer memory allocation.

The boot sequence can be commanded at any time by writing 1 to the WSEQ_BOOT_START bit.

Table 4-119. Write Sequencer Control—Boot Sequence

Register Address	Bit	Label	Default	Description
R24 (0x0018)	1	WSEQ_BOOT_		Writing 1 to this bit starts the write sequencer at the index location configured for
Write_Sequencer_ Ctrl 2		START		the Boot Sequence. The Boot Sequence start index is 384 (0x180).
				$\frac{1}{100}$

4.18.7 Sequencer Status Indication

The status of the write sequencer can be read using WSEQ_BUSY and WSEQ_CURRENT_INDEX, as described in Table 4-120. When the WSEQ_BUSY bit is asserted, this indicates that the write sequencer is busy.

The index address of the most recent write sequencer command can be read from the WSEQ_CURRENT_INDEX field. This can be used to provide a precise indication of the write sequencer progress.

Register Address	Bit	Label	Default	Description
R23 (0x0017)	9	WSEQ_BUSY	0	Sequencer Busy flag (Read Only).
Write_Sequencer_		(read only)		0 = Sequencer idle
Ctrl_1				1 = Sequencer busy
		WSEQ_CURRENT_ INDEX[8:0]		Sequence Current Index. This indicates the memory location of the most recently accessed command in the write sequencer memory.
		(read only)		Coding is the same as WSEQ_START_INDEX.

4.18.8 **Programming a Sequence**

A control-write sequence comprises a series of write operations to data bits within the control register map. Standard write operations are defined by five fields, contained within a single 32-bit register. An extended instruction set is also defined; the associated actions make use of alternate definitions of the 32-bit registers.



The sequencer instruction fields are replicated 508 times, defining each of the sequencer's 508 possible index addresses. Many sequences can be stored in the sequencer memory at the same time, with each assigned a unique range of index addresses. The WSEQ_DELAY*n* field is used to identify the end-of-sequence position, as described below.

The general definition of the sequencer instruction fields is described as follows, where *n* denotes the sequencer index address (valid from 0 to 507):

- WSEQ_DATA_WIDTH*n* is a 3-bit field that identifies the width of the data block to be written. Note that the maximum value of this field selects a width of 8 bits; writes to fields that are larger than 8 bits wide must be performed using two separate operations of the write sequencer.
- WSEQ_ADDRn is a 12-bit field containing the register address in which the data should be written. The applicable
 register address is referenced to the base address currently configured for the sequencer—it is calculated as: (base
 address * 512) + WSEQ_ADDRn. Note that the base address is configured using the sequencer's extended
 instruction set.
- WSEQ_DELAY*n* is a 4-bit field that controls the waiting time between the current step and the next step in the sequence (i.e., the delay occurs after the write in which it was called). The total delay time per step (including execution) is defined below, giving a useful range of execution/delay times from 3.3 µs up to 1 s per step.

If WSEQ_DELAYn = 0x0 or 0xF, the step execution time is 3.3 μ s

For all other values, the step execution time is 61.44 μ s x ((2 WSEQ_DELAY) – 1)

Setting this field to 0xF identifies the step as the last in the sequence

- WSEQ_DATA_START*n* is a 4-bit field that identifies the LSB position within the selected control register to which the data should be written. For example, setting WSEQ_DATA_START*n* = 0100 selects bit [4] as the LSB position of the data to be written.
- WSEQ_DATAn is an 8-bit field that contains the data to be written to the selected control register. The WSEQ_DATA_WIDTHn field determines how many of these bits are written to the selected control register; the most significant bits (above the number indicated by WSEQ_DATA_WIDTHn) are ignored.

The extended instruction set for the write sequencer is accessed by setting WSEQ_MODE*n* (bit [28]) in the respective sequencer definition register. The extended instruction set comprises the following functions:

- If bits [31:24] = 0x11, the register base address is set equal to the value contained in bits [23:0].
- If bits [31:16] = 0x12FF, the sequencer performs an unconditional jump to the index location defined in bits [15:0]. The index location is valid in the range 0 to 507 (0x1FB).
- All other settings within the extended instruction set are reserved.

The control field definitions for Step 0 are described in Table 4-121. The equivalent definitions also apply to Step 1 through Step 507, in the subsequent register address locations.



Register Address	Bit	Label	Default	Description
R12288 (0x3000)	31:29	WSEQ_DATA_	000	Width of the data block written in this sequence step.
WSEQ_		WIDTH0[2:0]		000 = 1 bit 011 = 4 bits 110 = 7 bits
Sequence_1				001 = 2 bits 100 = 5 bits 111 = 8 bits
				010 = 3 bits 101 = 6 bits
	28	WSEQ_MODE0	0	Extended Sequencer Instruction select
				0 = Basic instruction set
				1 = Extended instruction set
	27:16	WSEQ_ADDR0[11:0]	0x000	Control Register Address to be written to in this sequence step.
				The register address is calculated as: (Base Address * 512) + WSEQ_ADDRn.
				Base Address is 0x00_0000 by default, and is configured using the sequencer's
				extended instruction set.
	15:12	WSEQ_DELAY0[3:0]	0000	, , , , , , , , , , , , , , , , , , , ,
				0x0 = 3.3 μs
				0x1 to 0xE = 61.44 μs x ((2 ^{WSEQ_DELAY})–1)
				0xF = End of sequence marker
	11:8	WSEQ_DATA_	0000	Bit position of the LSB of the data block written in this sequence step.
		START0[3:0]		0000 = Bit 0
				1111 = Bit 15
	7:0	WSEQ_DATA0[7:0]	0x00	Data to be written in this sequence step. When the data width is less than 8 bits,
				one or more of the MSBs of WSEQ_DATA <i>n</i> are ignored. It is recommended that unused bits be cleared.
				unused bits be cleared.

Table 4-121. Write Sequencer Control—Programming a Sequence

4.18.9 Sequencer Memory Definition

The write sequencer memory defines up to 508 write operations; these are indexed as 0 to 507 in the sequencer memory map.

The write sequencer memory reverts to its default contents following power-on reset, a hardware reset, or a Sleep Mode transition. In these cases, the sequence memory contains the boot sequence and the OUT1–OUT3 signal path enable/ disable sequences; the remainder of the sequence memory is undefined.

User-defined sequences can be programmed after power-up. The user-defined control sequences must be reconfigured by the host processor following power-on reset, a hardware reset, or a Sleep Mode transition. Note that all control sequences are maintained in the sequencer memory through software reset. See Section 5.2 for a summary of the CS42L92 memory reset conditions.

The default control sequences can be overwritten in the sequencer memory, if required. Note that the headphone/earpiece output path enable bits (HPnx_ENA) always trigger the write sequencer (at the predetermined start index addresses).

Writing 1 to the WSEQ_LOAD_MEM bit clears the sequencer memory to the power-on reset state.

Register Address	Bit	Label	Default	Description
R24 (0x0018)	0	WSEQ_LOAD_	0	Writing 1 to this bit resets the sequencer memory to the power-on reset
Write_Sequencer_Ctrl_2		MEM		state.

The sequencer memory is summarized in Table 4-123. User-defined sequences should be assigned space within the allocated portion (user space) of the write sequencer memory.

The start index for the user-defined sequences is configured using the fields described in Table 4-114 through Table 4-118.



Table 4-123. Write Sequencer Memory Allocation

Description	Sequence Index Range
Default Sequences	0 to 302
User Space	303 to 383
Boot Sequence	384 to 507

4.19 Charge Pumps, Regulators, and Voltage Reference

The CS42L92 incorporates two charge-pump circuits and an LDO-regulator circuit to generate supply rails for internal functions and to support external microphone requirements. The CS42L92 also provides two MICBIAS generators (with six switchable outputs), which provide low noise reference voltages suitable for biasing ECM-type microphones or powering digital microphones.

In Sleep Mode, the CS42L92 can provide an unregulated voltage output that can be used to power an external microphone. See Section 4.13 for details of Sleep Mode.

The CPVDD domain (1.8 V) powers the Charge Pump 1 and Charge Pump 2 circuits. The CPVDD2 power domain (1.2 V) is an additional supply used by Charge Pump 1 only. Refer to Section 5.1 for recommended external components.

4.19.1 Charge Pump 1

Charge Pump 1 (CP1) is used to generate the positive and negative supply rails for the analog output drivers. CP1 is enabled automatically by the CS42L92 when required by the output drivers.

The Charge Pump 1 circuit is shown in Fig. 4-75.

4.19.2 Charge Pump 2 and LDO2 Regulator

Charge Pump 2 (CP2) powers LDO2, which provides the supply rail for analog input circuits and for the MICBIAS generators. CP2 and LDO2 are enabled by setting CP2_ENA.

If CP2 and LDO2 are enabled, the MICVDD voltage is selected using the LDO2_VSEL field. Note that, when one or more of the MICBIAS generators is operating in normal (regulator) mode, the MICVDD voltage must be at least 200 mV greater than the highest selected MICBIAS noutput voltages.

If CP2 and LDO2 are enabled, an internal bypass path may be selected, connecting the MICVDD pin directly to the CPVDD supply. This path is controlled using the CP2_BYPASS bit. Note that the bypass path is only supported when CP2 is enabled.

Note: The 32-kHz clock must be configured and enabled if CP2 is enabled in its normal operating mode. The 32-kHz clock is not required in bypass mode (CP2_BYPASS = 1). See Section 4.16 for details of the system clocks.

If CP2 is disabled, the CP2VOUT pin can be either floating or actively discharged. The behavior is configured using the CP2_DISCH bit.

If LDO2 is disabled, the MICVDD pin can be either floating or actively discharged. The behavior is configured using the LDO2_DISCH bit.

The MICVDD pin is connected to the output of LDO2. Note that the MICVDD does not support direct connection to an external supply; MICVDD is always powered internally to the CS42L92.

The Charge Pump 2 and LDO2 Regulator circuits are shown in Fig. 4-75. The associated control bits are described in Table 4-124.

Note that decoupling capacitors and flyback capacitors are required for these circuits. Refer to Section 5.1 for recommended external components.



4.19.3 Microphone Bias (MICBIAS) Control

There are two MICBIAS generators, which provide low-noise reference voltages suitable for biasing ECM-type microphones or powering digital microphones. Refer to Section 5.1.3 for recommended external components.

The MICBIAS generators are powered from MICVDD, which is generated by an internal charge pump and LDO, as shown in Fig. 4-75.

Switchable outputs from the MICBIAS generators allow six separate reference/supply outputs to be independently controlled. The MICBIAS regulators are enabled using the MICB1_ENA and MICB2_ENA bits. The MICBIAS output switches are enabled using the MICB1x_ENA and MICB2x_ENA (where *x* is A, B, C, or D).

The MICBIAS1 generator supports four switchable outputs (MICBIAS1A–MICBIAS1D). The MICBIAS2 generator supports two switchable outputs (MICBIAS2A–MICBIAS2B).

Note that, to enable any of the MICBIAS *nx* outputs, both the output switch and the respective regulator must be enabled.

When a MICBIAS output is disabled, it can be configured to be floating or to be actively discharged. This is configured using the MICB*n*_DISCH bits (for the MICBIAS regulators), and the MICB*nx*_DISCH bits (for the switched outputs). Each discharge path is only effective when the respective regulator, or switched output, is disabled.

The MICBIAS generators can each operate in Regulator Mode or in Bypass Mode. The applicable mode is selected using the MICB*n*_BYPASS bits.

In Regulator Mode (MICB $n_BYPASS = 0$), the output voltage is selected using the MICB n_LVL fields. In this mode, MICVDD must be at least 200 mV greater than the required MICBIAS output voltages. The MICBIAS outputs are powered from the MICVDD pin and use the internal band-gap circuit as a reference.

In Regulator Mode, the MICBIAS regulators are designed to operate without external decoupling capacitors. The regulators can be configured to support a capacitive load if required, using the MICB*n*_EXT_CAP bits. (This may be appropriate for a DMIC supply.) It is important that the external capacitance is compatible with the applicable MICB*n*_EXT_CAP setting. The compatible load conditions are detailed in Table 3-11.

In Bypass Mode (MICB $n_BYPASS = 1$), the respective outputs (MICBIASnx), when enabled, are connected directly to MICVDD. This enables a low power operating state. Note that the MICB n_EXT_CAP settings are not applicable in Bypass Mode—there are no restrictions on the external MICBIAS capacitance in Bypass Mode.

The MICBIAS generators incorporate a pop-free control circuit to ensure smooth transitions when the MICBIAS outputs are enabled or disabled in Bypass Mode; this feature is enabled using the MICB*n*_RATE bits.

The MICBIAS generators are shown in Fig. 4-75. The MICBIAS control fields are described in Table 4-124.

The maximum output current for each MICBIAS regulator is noted in Table 3-11. This limit must be observed for each set of MICBIAS *nx* outputs, especially if more than one microphone is connected to a single regulator. Note that the maximum output current differs between Regulator Mode and Bypass Mode.

4.19.3.1 MICBIAS output in Sleep Mode

The CS42L92 supports a low-power Sleep Mode, in which most functions are disabled and power consumption is minimized. The CS42L92 can maintain an unregulated voltage output in Sleep Mode, suitable for powering an external microphone in always-on applications. The INnx/DMICx connections are isolated from the CS42L92 circuits in Sleep Mode, to allow any connected microphone to be used by another circuit while Sleep Mode is selected.

If the MICB1A_AOD_ENA bit is set, the normal register-field configuration of CP2, LDO2, and MICBIAS1 are overridden. Under these conditions, the CP2, LDO2 and MICBIAS1 circuits are bypassed and the MICBIAS1A output is connected directly to the CPVDD supply.

The unregulated MICBIAS1A output can be maintained in Sleep Mode by setting MICB1A_AOD_ENA before the DCVDD supply is removed.



To minimize any transient effects during the Sleep Mode transition, the MICBIAS1A should be configured in bypass mode before Sleep Mode is enabled. The following control sequence is recommended:

- CP_BYPASS = 1
- MICB1_BYPASS = 1
- MICB1A_AOD_ENA = 1
- Remove DCVDD

To minimize any transient effects following a wake-up transition, the MICBIAS1A should be configured in bypass mode before clearing MICB1A_AOD_ENA. The following control sequence is recommended:

- Apply DCVDD
- LDO2_VSEL = (set as required)
- CP2_BYPASS = 0
- MICB1_LVL = (set as required)
- MICB1_BYPASS = 0
- MICB1_ENA = 1
- MICB1A_ENA = 1
- MICB1A_AOD_ENA = 0

4.19.4 Voltage-Reference Circuit

The CS42L92 incorporates a voltage-reference circuit, powered by AVDD. This circuit ensures the accuracy of the LDO-regulator and MICBIAS voltage settings.

4.19.5 Block Diagram and Control Registers

The charge-pump and regulator circuits are shown in Fig. 4-75. Note that decoupling capacitors and flyback capacitors are required for these circuits. Refer to Section 5.1 for recommended external components.



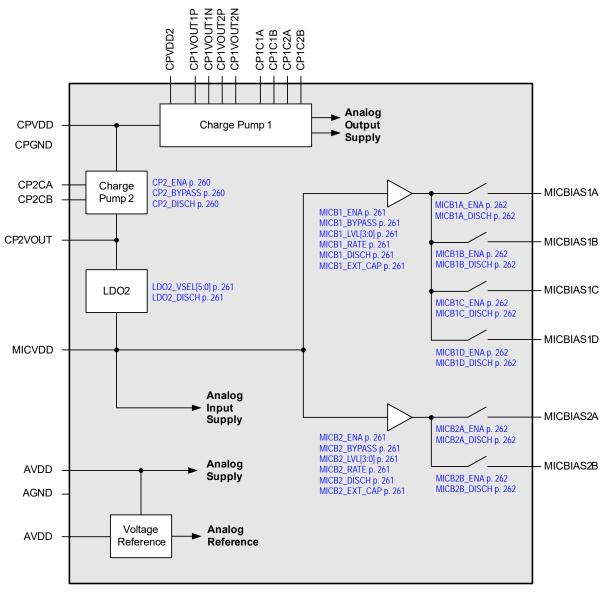


Figure 4-75. Charge Pumps and Regulators

The charge-pump and regulator control registers are described in Table 4-124.

Table 4-124.	Charge-Pump and LDO Control Registers
--------------	---------------------------------------

Register Address	Bit	Label	Default	Description	
R512 (0x0200)	2	CP2_DISCH	1	Charge Pump 2 Discharge	
Mic_Charge_				0 = CP2VOUT floating when disabled	
Pump_1				1 = CP2VOUT discharged when disabled	
	1	CP2_BYPASS	1	Charge Pump 2 and LDO2 Bypass Mode	
				0 = Normal	
				1 = Bypass Mode	
				In Bypass Mode, CPVDD is connected directly to MICVDD.	
				Note that CP2_ENA must also be set.	
	0	CP2_ENA	1	Charge Pump 2 and LDO2 Control	
				(Provides analog input and MICVDD supplies)	
				0 = Disabled	
				1 = Enabled	



Register Address	Bit	Label	Default	Description
R531 (0x0213)		LDO2_VSEL[5:0]	0x1F	LDO2 Output Voltage Select 1
LDO2_Control_1				0x00 = 0.900 V 0x13 = 1.375 V (100-mV steps)
				0x01 = 0.925 V 0x14 = 1.400 V 0x26 = 3.200 V
				0x02 = 0.950 V 0x15 = 1.500 V 0x27 to 0x3F = 3.300 V
				(25-mV steps) 0x16 = 1.600 V
	2	LDO2_DISCH	1	LDO2 Discharge
		_		0 = MICVDD floating when disabled
				1 = MICVDD discharged when disabled
R536 (0x0218)	15	MICB1 EXT CAP	0	Microphone Bias 1 External Capacitor (when MICB1_BYPASS = 0).
Mic_Bias_Ctrl_1				Configures the MICBIAS1 regulator according to the specified capacitance connected
				to the MICBIAS1x outputs.
				0 = No external capacitor
				1 = External capacitor connected
	8:5	MICB1_LVL[3:0]	0x7	Microphone Bias 1 Voltage Control (when MICB1_BYPASS = 0)
				0x0 = 1.5 V (0.1-V steps) 0xD to 0xF = 2.8 V
				0x1 = 1.6 V 0xC = 2.7 V
	3	MICB1_RATE	0	Microphone Bias 1 Rate (Bypass Mode)
				0 = Fast start-up/shutdown
				1 = Pop-free start-up/shutdown
	2	MICB1_DISCH	1	Microphone Bias 1 Discharge
				0 = MICBIAS1 floating when disabled
				1 = MICBIAS1 discharged when disabled
	1	MICB1_BYPASS	1	Microphone Bias 1 Mode
				0 = Regulator Mode
				1 = Bypass Mode
	0	MICB1_ENA	0	Microphone Bias 1 Enable
				0 = Disabled
				1 = Enabled
		Microphone Bias 2 External Capacitor (when MICB2_BYPASS = 0). Configures the MICBIAS2 regulator according to the specified capacitance connected to the		
				MICBIAS2x outputs.
				0 = No external capacitor
				1 = External capacitor connected
	8:5	MICB2_LVL[3:0]	0x7	Microphone Bias 2 Voltage Control (when MICB2_BYPASS = 0)
				0x0 = 1.5 V (0.1-V steps) 0xD to 0xF = 2.8 V
				0x1 = 1.6 V 0xC = 2.7 V
	3	MICB2_RATE	0	Microphone Bias 2 Rate (Bypass Mode)
		_		0 = Fast start-up/shutdown
				1 = Pop-free start-up/shutdown
	2	MICB2_DISCH	1	Microphone Bias 2 Discharge
				0 = MICBIAS2 floating when disabled
				1 = MICBIAS2 discharged when disabled
	1	MICB2_BYPASS	1	Microphone Bias 2 Mode
				0 = Regulator Mode
				1 = Bypass Mode
	0	MICB2_ENA	0	Microphone Bias 2 Enable
		_		0 = Disabled
				1 = Enabled

Table 4-124. Charge-Pump and LDO Control Registers (Cont.)



Register Address	Bit	Label	Default	Description
R540 (0x021C)	13	MICB1D_DISCH	1	Microphone Bias 1D Discharge
Mic_Bias_Ctrl_5				0 = MICBIAS1D floating when disabled
				1 = MICBIAS1D discharged when disabled
	12	MICB1D_ENA	0	Microphone Bias 1D Enable
				0 = Disabled
				1 = Enabled
	9	MICB1C_DISCH	1	Microphone Bias 1C Discharge
				0 = MICBIAS1C floating when disabled
				1 = MICBIAS1C discharged when disabled
	8	MICB1C_ENA	0	Microphone Bias 1C Enable
				0 = Disabled
				1 = Enabled
	5	MICB1B_DISCH	1	Microphone Bias 1B Discharge
				0 = MICBIAS1B floating when disabled
				1 = MICBIAS1B discharged when disabled
	4	MICB1B_ENA	0	Microphone Bias 1B Enable
				0 = Disabled
				1 = Enabled
	1	MICB1A_DISCH	1	Microphone Bias 1A Discharge
				0 = MICBIAS1A floating when disabled
				1 = MICBIAS1A discharged when disabled
	0	MICB1A_ENA	0	Microphone Bias 1A Enable
				0 = Disabled
				1 = Enabled
R542 (0x021E)	5	MICB2B_DISCH	1	Microphone Bias 2B Discharge
Mic_Bias_Ctrl_6				0 = MICBIAS2B floating when disabled
				1 = MICBIAS2B discharged when disabled
	4	MICB2B_ENA	0	Microphone Bias 2B Enable
				0 = Disabled
				1 = Enabled
	1	MICB2A_DISCH	1	Microphone Bias 2A Discharge
				0 = MICBIAS2A floating when disabled
				1 = MICBIAS2A discharged when disabled
	0	MICB2A_ENA	0	Microphone Bias 2A Enable
				0 = Disabled
				1 = Enabled
R723 (0x02D3)	14	MICB1A_AOD_	0	Microphone Bias 1A Always-On Enable
Jack_detect_		ENA		0 = Disabled
analogue				1 = Enabled

Table 4-124. Charge-Pump and LDO Control Registers (Cont.)

1.See Table 4-125 for LDO2 output voltage definition.

Table 4-125 lists the LDO2 voltage control settings.

LDO2_VSEL[5:0]	LDO Output	LDO2_VSEL[5:0]	LDO Output
0x00	0.900 V	0x15	1.500 V
0x01	0.925 V	0x16	1.600 V
0x02	0.950 V	0x17	1.700 V
0x03	0.975 V	0x18	1.800 V
0x04	1.000 V	0x19	1.900 V
0x05	1.025 V	0x1A	2.000 V
0x06	1.050 V	0x1B	2.100 V
0x07	1.075 V	0x1C	2.200 V
0x08	1.100 V	0x1D	2.300 V
0x09	1.125 V	0x1E	2.400V
0x0A	1.150 V	0x1F	2.500 V

Table 4-125. LDO2 Voltage Control



LDO2_VSEL[5:0]	LDO Output	LDO2_VSEL[5:0]	LDO Output
0x0B	1.175 V	0x20	2.600 V
0x0C	1.200 V	0x21	2.700 V
0x0D	1.225 V	0x22	2.800 V
0x0E	1.250 V	0x23	2.900 V
0x0F	1.275 V	0x24	3.000 V
0x10	1.300 V	0x25	3.100 V
0x11	1.325 V	0x26	3.200 V
0x12	1.350 V	0x27	3.300 V
0x13	1.375 V	0x28 to 0x3F	3.300 V
0x14	1.400 V		

Table 4-125. LDO2 Voltage Control (Cont.)

4.20 JTAG Interface

The JTAG interface provides test and debug access to the CS42L92. The interface comprises five connections, some of which are multiplexed with the GPIO13–GPIO15 pins, as noted in Table 4-126.

Pin No	Pin Name	JTAG Function	JTAG Description
N8	AIF3BCLK/GPIO14	TCK	Clock input
R8	AIF3RXDAT/GPIO15	TDI	Data input
F7	TDO	TDO	Data output
J8	AIF3TXDAT/GPIO13	TMS	Mode select input
D7	TRST	TRST	Test access port reset input (active low)

Table 4-126. JTAG Interface Connections

The JTAG interface is supported whenever the GPIO13–GPIO15 are configured as GPIO inputs. To allow JTAG operation, the following conditions must be met:

- GPn_FN = 0x001 (n = 13–15)
- GP*n*_DIR = 0x1 (*n* = 13–15)

The GPn_FN and GPn_DIR fields are defined in Table 4-93. Note that the above conditions are also the default values of these control fields.

For normal operation (test and debug access disabled), the JTAG interface should be held in reset. An internal pull-down resistor holds the TRST pin low (i.e., JTAG interface is held in reset) when not actively driven. It is recommended to connect the TRST pin to DGND, if the JTAG interface function is not required.

Integrated pull-up and pull-down resistors can be enabled on the TCK, TDI, and TMS pins. This is provided as part of the GPIO functionality, and provides a flexible capability for interfacing with other devices. The pull-up and pull-down resistors can be configured independently using the fields described in Table 4-93.

If the JTAG interface is enabled (TRST deasserted and TCK active) at the time of any reset, a software reset must be scheduled, with the TCK input stopped or TRST asserted (Logic 0), before using the JTAG interface.

It is recommended to always schedule a software reset before starting the JTAG clock or deasserting the JTAG reset. In this event, the JTAG interface should be held in its reset state until the software reset has completed, and the BOOT_DONE_STSx bits have been set. See Section 4.23 for further details of the CS42L92 software reset.

4.21 Short-Circuit Protection

The CS42L92 provides short-circuit protection on the headphone output drivers.

The short-circuit protection function for the headphone output paths operates continuously if the respective output driver is enabled. If a short circuit is detected on the headphone output, current limiting is applied to protect the respective output driver. Note that the driver continues to operate, but the output is current-limited.



The headphone short-circuit protection function provides input to the interrupt control circuit and can be used to trigger an interrupt event when a short-circuit condition is detected; see Section 4.15.

4.22 Power-On Reset (POR)

The CS42L92 remains in the reset state until AVDD, DBVDD, and DCVDD are above their respective reset thresholds. Note that specified device performance is not assured outside the voltage ranges defined in Table 3-3.

After the initial power-up, the POR is rescheduled following an interruption to the DBVDD or AVDD supplies.

If the CS42L92 SLIMbus component is in its operational state, it must be reset before scheduling a POR. See Section 4.10 for details of the SLIMbus reset control messages.

4.22.1 Boot Sequence

Following power-on reset, a boot sequence is executed. The BOOT_DONE_STS*x* bits are asserted on completion of the boot sequence, as described in Table 4-127. Control-register writes should not be attempted until BOOT_DONE_STS*x* has been asserted. Note that the BOOT_DONE_STS1 and BOOT_DONE_STS2 bits provide the same information.

The BOOT_DONE_STS*x* signal is an input to the interrupt control circuit and can be used to trigger an interrupt event on completion of the boot sequence; see Section 4.15. Under default register conditions, a falling edge on the IRQ pin indicates completion of the boot sequence.

For details of the boot sequence, see Section 4.18.

Register Address	Bit	Label	Default	Description
R6272 (0x1880)	7	BOOT_DONE_	0	Boot Status
IRQ1_Raw_		STS1		0 = Busy (boot sequence in progress)
Status_1				1 = Idle (boot sequence completed)
				Control register writes should not be attempted until Boot Sequence has completed.
R6528 (0x1980)	7	BOOT_DONE_	0	Boot Status
IRQ2_Raw_		STS2		0 = Busy (boot sequence in progress)
Status_1				1 = Idle (boot sequence completed)
				Control register writes should not be attempted until Boot Sequence has completed.

Table 4-127. Device Boot-Up Status

4.22.2 Digital I/O Status in Reset

Table 1-1 describes the default status of the CS42L92 digital I/O pins on completion of power-on reset, prior to any register writes. The same default conditions are also applicable on completion of a hardware reset or software reset (see Section 4.23).

The same default conditions are applicable following a wake-up transition, except for the IRQ and RESET pins. These are always-on pins whose configuration is unchanged in Sleep Mode and during a wake-up transition.

Note that the default conditions described in Table 1-1 are not valid if modified by the boot sequence or by a wake-up control sequence. See Section 4.18 for details of these functions.

4.23 Hardware Reset, Software Reset, Wake-Up, and Device ID

The CS42L92 supports hardware- and software-controlled reset functions. The reset functions, and the Sleep/Wake-Up state transitions, provide similar (but not identical) functionality. Each of these is described in the following subsections.

The CS42L92 device ID can be read from the Software_Reset (R0) control register, as described in Section 4.23.7.



4.23.1 Hardware Reset

The CS42L92 provides a hardware reset function, which is executed whenever the RESET input is asserted (Logic 0). The RESET input is active low and is referenced to the DBVDD power domain. A hardware reset causes all of the CS42L92 control registers to be reset to their default states.

An internal pull-up resistor is enabled by default on the RESET pin; this can be configured using the RESET_PU bit. A pull-down resistor is also available, as described in Table 4-128. When the pull-up and pull-down resistors are both enabled, the CS42L92 provides a bus keeper function on the RESET pin. The bus keeper function holds the input logic level unchanged whenever the external circuit removes the drive (e.g., if the signal is tristated).

If the CS42L92 SLIMbus component is in its operational state, it must be reset prior to scheduling a hardware reset. See Section 4.10 for details of the SLIMbus reset control messages.

Register Address	Bit	Label	Default	Description
R6864 (0x1AD0)	1	RESET_PU	1	RESET Pull-up enable
AOD_Pad_Ctrl				0 = Disabled
				1 = Enabled
				Note: If RESET_PD and RESET_PU are both set, a bus keeper function is enabled on the RESET pin.
	0	RESET_PD	0	RESET Pull-down enable
				0 = Disabled
				1 = Enabled
				Not<u>e:</u> If RESET_PD and RESET_PU are both set, a bus keeper function is enabled on the RESET pin.

4.23.2 Software Reset

A software reset is executed by writing any value to register R0. A software reset causes most of the CS42L92 control registers to be reset to their default states. Note that the control-write sequencer memory is retained during software reset.

Note that the first register read/write operation following a software reset may be unsuccessful, if the register access is attempted via a different control interface to the one that commanded the software reset. Note that only the first register read/write is affected, and only when using more than one control interface.

4.23.3 Wake-Up

The CS42L92 is in Sleep Mode when AVDD and DBVDD are present, and DCVDD is below its reset threshold. (Note that specific control requirements are also applicable for entering Sleep Mode, as described in Section 4.13.)

In Sleep Mode, most of the digital core (and control registers) are held in reset; selected functions and control registers are maintained via an always-on internal supply domain. See Section 4.13 for details of the always-on functions.

A wake-up transition (from Sleep Mode) is similar to a software reset, but selected functions and control registers are maintained via an always-on internal supply domain—the always-on registers are not reset during wake-up. See Section 4.13 for details of the always-on functions.

4.23.4 Write Sequencer and DSP Firmware Memory Control in Reset and Wake-Up

The control-write sequencer memory contents reverts to its default contents following power-on reset, a hardware reset, or a Sleep Mode transition. The control sequences (including any user-defined sequences) are maintained in the sequencer memory through software reset.

The DSP firmware memory contents are cleared following power-on reset, a hardware reset, or a Sleep Mode transition. The firmware memory contents are not affected by software reset, provided DCVDD is held above its reset threshold.

See Section 5.2 for a summary of the CS42L92 memory reset conditions.



4.23.5 Boot Sequence

Following hardware reset, software reset, or wake-up from Sleep Mode, a boot sequence is executed. The BOOT_DONE_ STSx bits (see Table 4-127) are deasserted during hardware reset and software reset, and also in Sleep Mode. The BOOT_DONE_STSx bits are asserted on completion of the boot sequence. Control register writes should not be attempted until BOOT_DONE_STSx has been asserted.

The BOOT_DONE_STSx status is an input to the interrupt control circuit and can be used to trigger an interrupt event; see Section 4.15. Note that the BOOT_DONE_STS1 and BOOT_DONE_STS2 bits provide the same information.

For details of the boot sequence, see Section 4.18.

4.23.6 Digital I/O Status in Reset

The status of the CS42L92 digital I/O pins following hardware reset, software reset, or wake-up is described in Section 4.22.2.

4.23.7 Device ID

The device ID can be read from Register R0. The hardware revision can be read from Register R1.

The software revision can be read from Register R2. The software revision code is incremented if software driver compatibility or software feature support is changed.

Register Address	Bit	Label	Default	Description
R0 (0x0000)		SW_RST_DEV_	0x6371	Writing to this register resets all registers to their default state.
Software_Reset		ID[15:0]		Reading from this register indicates Device ID 0x6371.
R1 (0x0001)	7:0	HW_	—	Hardware Device revision.
Hardware_ Revision		REVISION[7:0]		This field is incremented for every new revision of the device.
R2 (0x0002)	7:0	SW_	—	Software Device revision.
Software_Revision		REVISION[7:0]		This field is incremented if software driver compatibility or software feature support is changed.

Table 4-129. Device Reset and ID



5 Applications

5.1 Recommended External Components

This section provides information on the recommended external components for use with the CS42L92.

5.1.1 Analog Input Paths

The CS42L92 supports up to eight analog audio input connections. Each input is biased to the internal DC reference, VREF. (Note that this reference voltage is present on the VREFC pin.) A DC-blocking capacitor is required for each analog input pin used in the target application. The choice of capacitor is determined by the filter that is formed between that capacitor and the impedance of the input pin. The circuit is shown in Fig. 5-1.

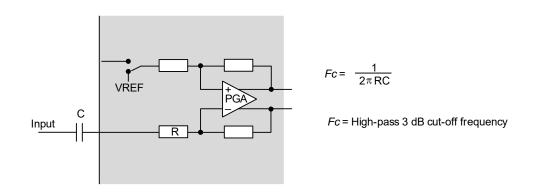


Figure 5-1. Audio Input Path DC-Blocking Capacitor

In accordance with the CS42L92 input pin resistance (see Table 3-5), a $1-\mu$ F capacitance for all input connections gives good results in most cases, with a 3-dB cut-off frequency around 13 Hz.

Ceramic capacitors are suitable, but take care to ensure the desired capacitance is maintained at the AVDD operating voltage. Also, ceramic capacitors may show microphonic effects, where vibrations and mechanical conditions give rise to electrical signals. This is particularly problematic for microphone input paths where a large signal gain is required.

A single capacitor is required for a single-ended line or microphone input connection. For a differential input connection, a DC-blocking capacitor is required on both input pins.

The external connections for single-ended and differential microphones, incorporating the CS42L92 microphone bias circuit, are shown in Fig. 5-2.

5.1.2 DMIC Input Paths

The CS42L92 supports up to eight channels of DMIC input; two channels of audio data can be multiplexed on each DMICDAT*n* pin. Each stereo pair is clocked using the respective DMICCLK*n* pin.

The external connections for digital microphones, incorporating the CS42L92 microphone bias circuit, are shown in Fig. 5-4. Ceramic decoupling capacitors for the digital microphones may be required—refer to the specific recommendations for the application microphones.

If two microphones are connected to a single DMICDAT pin, the microphones must be configured to ensure that the Left mic transmits a data bit when DMICCLK is high, and the Right mic transmits a data bit when DMICCLK is low. The CS42L92 samples the DMIC data at the end of each DMICCLK phase. Each microphone must tristate its data output when the other microphone is transmitting. Integrated pull-down resistors can be enabled on the DMICDAT pins if required.

The voltage reference for each DMIC interface is selectable. It is important that the selected reference for the CS42L92 interface is compatible with the applicable configuration of the external microphone.



5.1.3 Microphone Bias Circuit

The CS42L92 is designed to interface easily with analog or digital microphones.

Each microphone requires a bias current (electret condenser microphones) or voltage supply (silicon microphones); these can be provided by the MICBIAS regulators on the CS42L92. Two MICBIAS generators are available; switchable outputs allow six separate reference/supply outputs to be independently controlled.

Note that the MICVDD pin can also be used (instead of MICBIAS*nx*) as a reference or power supply for external microphones. The MICBIAS outputs are recommended, as these offer better noise performance and independent enable/ disable control.

Analog microphones may be connected in single-ended or differential configurations, as shown in Fig. 5-2. The differential configuration provides better performance due to its rejection of common-mode noise; the single-ended method provides a reduction in external component count.

A bias resistor is required when using an ECM. The bias resistor should be chosen according to the minimum operating impedance of the microphone and MICBIAS voltage so that the maximum bias current of the CS42L92 is not exceeded.

A 2.2-k Ω bias resistor is recommended; this provides compatibility with a wide range of microphone components.

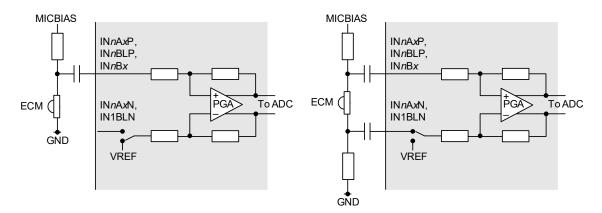


Figure 5-2. Single-Ended and Differential Analog Microphone Connections

Analog MEMS microphones can be connected to the CS42L92 as shown in Fig. 5-3. In this configuration, the MICBIAS generators provide a low-noise supply for the microphones; a bias resistor is not required.

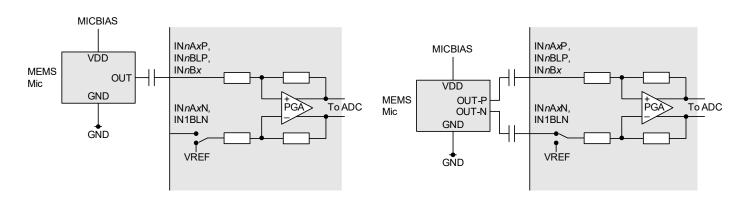


Figure 5-3. Single-Ended and Differential Analog Microphone Connections

DMIC connection to the CS42L92 is shown in Fig. 5-4. Note that ceramic decoupling capacitors at the DMIC power supply pins may be required—refer to the specific recommendations for the application microphones.



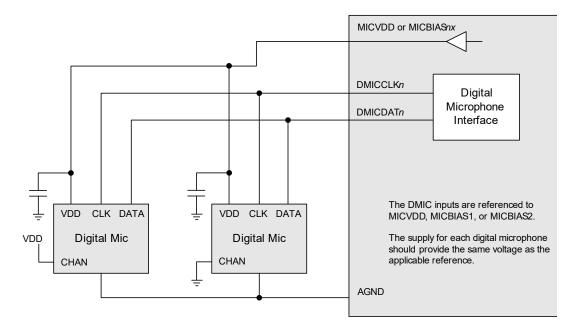


Figure 5-4. DMIC Connection

Each MICBIAS generator can operate in Regulator Mode or in Bypass Mode. See Section 4.19 for details of the MICBIAS generators.

In Regulator Mode, the MICBIAS regulators are designed to operate without external decoupling capacitors. The regulators can be configured to support a capacitive load if required (e.g., for DMIC supply decoupling). The compatible load conditions are detailed in Table 3-11.

If the capacitive load on MICBIAS1 or MICBIAS2 exceeds the specified conditions for Regulator Mode (e.g., due to a decoupling capacitor or long PCB trace), the respective generator must be configured in Bypass Mode.

The maximum output current for each MICBIAS regulator is noted in Table 3-11. This limit must be observed for each set of MICBIAS *nx* outputs, especially if more than one microphone is connected to a single regulator. Note that the maximum output current differs between Regulator Mode and Bypass Mode. The MICBIAS output voltage can be adjusted using register control in Regulator Mode.

5.1.4 Headphone Driver Output Path

The CS42L92 provides four stereo headphone output drivers. These outputs are all ground referenced, allowing direct connection to the external loads. There is no requirement for DC-blocking capacitors.

In single-ended (default) configuration, the headphone outputs comprise eight independently controlled output channels, for up to four stereo headphone or line outputs. In mono (BTL) mode, the headphone drivers support up to four differential outputs, suitable for a mono earpiece or hearing coil load.

Note that the OUT3 signal path is common to the HPOUT3 and HPOUT4 drivers—only one of these stereo drivers may be enabled at any time.

The headphone outputs incorporate a common mode, or ground loop, feedback path that provides rejection of system-related ground noise. The feedback pins must be connected to ground for normal operation of the headphone outputs. Five feedback pins are provided (HPOUTFB1–HPOUTFB5). The ground feedback path for each HPOUT path is selected using the HP*n*_GND_SEL bits as follows:

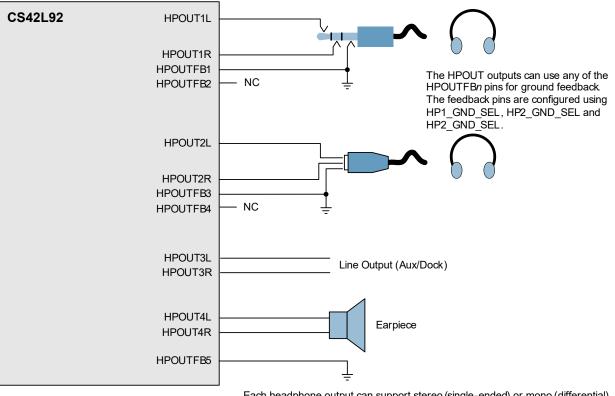
- The ground feedback path for HPOUT1 and HPOUT2 headphone outputs is selected using the HP1_GND_SEL and HP2_GND_SEL register fields respectively—see Table 4-81.
- The ground feedback path for HPOUT3 and HPOUT4 headphone outputs is selected using the HP3_GND_SEL field.



The selected feedback pin should be connected to GND as close as possible to the respective headphone jack ground pin, as shown in Fig. 5-5. In mono (differential) mode, the feedback pins should be connected to the ground plane that is closest to the earpiece output PCB tracks.

It is recommended to ensure that the electrical characteristics of the PCB traces for each output pair are closely matched. This is particularly important to matching the two traces of a differential (BTL) output.

Typical headphone and earpiece connections are shown in Fig. 5-5.



Each headphone output can support stereo (single-ended) or mono (differential) output. The illustration shows the configuration for a typical application .

Figure 5-5. Headphone and Earpiece Connection

It is common for ESD diodes to be wired to pins that link to external connectors. This provides protection from potentially harmful ESD effects. In a typical application, ESD diodes would be recommended if the headphone paths (HPOUT1–HPOUT4) are used for external headphone or line output.

The HPOUT*n* outputs are ground-referenced, and the respective voltages may swing between +1.8V and –1.8V. The ESD diode configuration must be carefully chosen.

The recommended ESD diode configuration for these ground-referenced outputs is shown in Fig. 5-6. The back-to-back arrangement prevents clipping and distortion of the output signal.

Note that similar care is required when connecting the CS42L92 outputs to external circuits that provide input path ESD protection; the configuration on those input circuits must be correctly designed to accommodate ground-referenced signals.



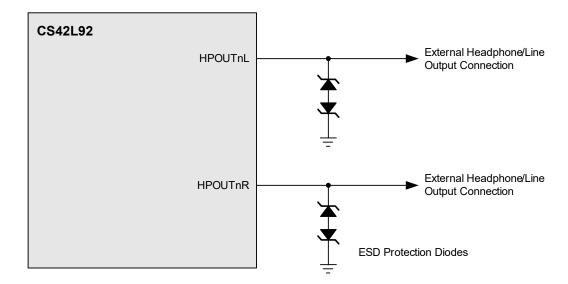


Figure 5-6. ESD Diode Configuration for External Output Connections

5.1.5 Power Supply/Reference Decoupling

Electrical coupling exists particularly in digital logic systems where switching in one subsystem causes fluctuations on the power supply. This effect occurs because the inductance of the power supply acts in opposition to the changes in current flow that are caused by the logic switching. The resultant variations (spikes) in the power-supply voltage can cause malfunctions and unintentional behavior in other components. A decoupling (bypass) capacitor can be used as an energy storage component that provides power to the decoupled circuit for the duration of these power-supply variations, protecting it from malfunctions that could otherwise arise.

Coupling also occurs in a lower frequency form when ripple is present on the power supply rail caused by changes in the load current or by limitations of the power-supply regulation method. In audio components such as the CS42L92, these variations can alter the performance of the signal path, leading to degradation in signal quality. A decoupling capacitor can be used to filter these effects by presenting the ripple voltage with a low-impedance path that does not affect the circuit to be decoupled.

These coupling effects are addressed by placing a capacitor between the supply rail and the corresponding ground reference. In the case of systems comprising multiple power supply rails, decoupling should be provided on each rail.

PCB layout is also a contributory factor for coupling effects. If multiple power supply rails are connected to a single supply source, it is recommended to provide separate PCB tracks connecting each rail to the supply. See Section 5.5 for PCB-layout recommendations.

The recommended power-supply decoupling capacitors for CS42L92 are detailed in Table 5-1.

Power Supply	Decoupling Capacitor
AVDD1, AVDD2	2 x 1.0 μ F ceramic—one capacitor on each AVDD <i>n</i> pin
CPVDD1	4.7 μF ceramic
CPVDD2	4.7 μF ceramic
DBVDD	1 x 0.1 μF ceramic ¹
DCVDD	$2 \text{ x} 1.0 \ \mu\text{F}$ ceramic—one capacitor on each DCVDD pin
FLLVDD	4.7 μF ceramic
MICVDD	4.7 μF ceramic
VREFC	2.2 μF ceramic

Table 5-1. Power Supply Decoupling Capacitors

1. Total capacitance of 4.7 μ F is required for the DBVDD domain. This can be provided by dedicated DBVDD decoupling or by other capacitors on the same power rail.



All decoupling capacitors should be placed as close as possible to the CS42L92 device. The connection between AGND, the AVDD decoupling capacitor, and the main system ground should be made at a single point as close as possible to the AGND balls of the CS42L92.

Due to the wide tolerance of many types of ceramic capacitors, care must be taken to ensure that the selected components provide the required capacitance across the required temperature and voltage ranges in the intended application. For most application the use of ceramic capacitors with capacitor dielectric X7R is recommended.

5.1.6 Charge-Pump Components

The CS42L92 incorporates two charge-pump circuits (CP1 and CP2).

CP1 generates the CP1VOUT*nx* supply rails for the ground-referenced headphone drivers; CP2 generates the CP2VOUT supply rail for the microphone bias (MICBIAS) regulators.

Decoupling capacitors are required on each of the charge-pump outputs. Two fly-back capacitors are required for CP1; a single fly-back capacitor is required for CP2.

The recommended charge-pump capacitors for CS42L92 are detailed in Table 5-2.

Description	Capacitor
CP1VOUT1P decoupling	Required capacitance is 2.0 µF at 2 V. Suitable component typically 4.7 µF.
CP1VOUT1N decoupling	Required capacitance is 2.0 μF at 2 V. Suitable component typically 4.7 μF.
CP1 fly-back 1 (connect between CP1C1A and CP1C1B)	Required capacitance is 1.0 µF at 2 V. Suitable component typically 2.2 µF.
CP1VOUT2P decoupling	Required capacitance is 2.0 µF at 2 V. Suitable component typically 4.7 µF.
CP1VOUT2N decoupling	Required capacitance is 2.0 µF at 2 V. Suitable component typically 4.7 µF.
CP1 fly-back 2 (connect between CP1C2A and CP1C2B)	Required capacitance is 1.0 µF at 2 V. Suitable component typically 2.2 µF.
CP2VOUT decoupling	Required capacitance is 1.0 μF at 3.6 V. Suitable component typically 4.7 μF.
CP2 fly-back (connect between CP2CA and CP2CB)	Required capacitance is 220 nF at 2 V. Suitable component typically 470 nF.

Table 5-2. Charge-Pump External Capacitors

Ceramic capacitors are recommended for these charge-pump requirements. Note that, due to the wide tolerance of many types of ceramic capacitors, care must be taken to ensure that the selected components provide the required capacitance across the required temperature and voltage ranges in the intended application. Ceramic capacitors with X7R dielectric are recommended.

The positioning of the charge-pump capacitors is important. These capacitors (particularly the fly-back capacitors) must be placed as close as possible to the CS42L92. The component choice and positioning of the CP1 components are more critical than those of CP2, due to the higher output power requirements of CP1.

5.1.7 External Accessory Detection Components

The external accessory detection circuit measures jack insertion using the JACKDET*n* pins. The insertion switch status is detected using an internal pull-up resistor circuit on the respective pin. Note that the logic thresholds associated with the JACKDET*n* pins differ from each other, as described in Table 3-11—this provides support for different jack switch configurations.

Microphone detection and key-button press detection is supported using the MICDET*n* pins. The applicable pin should be connected to one of the MICBIAS*nx* outputs, via a 2.2-k Ω bias resistor, as described in Section 5.1.3. Note that, when using the external accessory detection function, the MICBIAS*nx* resistor must be 2.2 k Ω ±2%.

A recommended circuit configuration, including headphone output on HPOUT1 and microphone connections, is shown in Fig. 5-7. See Section 5.1.1 for details of the DC-blocking microphone input capacitor selection.



The recommended external components and connections for microphone/push-button detection are shown in Fig. 5-7.

Note that, when using the microphone detect circuit, it is recommended to use the IN1AL*x*, IN1BL*x*, or IN1BR analog microphone input paths to ensure best immunity to electrical transients arising from the external accessory.

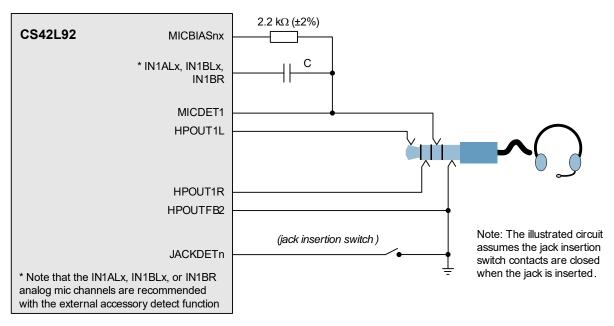


Figure 5-7. External Accessory Detection

The accessory detection circuit measures the impedance of an external load connected to one of the MICDET pins.

The microphone-detection circuit uses MICVDD, or any one of the MICBIAS*nx* sources, as a reference. The applicable source is configured using MICD*n*_BIAS_SRC.

With default register configuration, the CS42L92 can detect the presence of a typical microphone and up to four push buttons, using the components shown in Fig. 5-8. When the microphone detection circuit is enabled, each of the push buttons shown causes a different bit in the MICD*n*_LVL field to be set.

The choice of external resistor values must take into account the impedance of the microphone—the detected impedance corresponds to the combined parallel resistance of the microphone and any asserted push button. The components shown in Fig. 5-8 are examples only, assuming default impedance measurement ranges and a microphone impedance of 1 k Ω or higher.



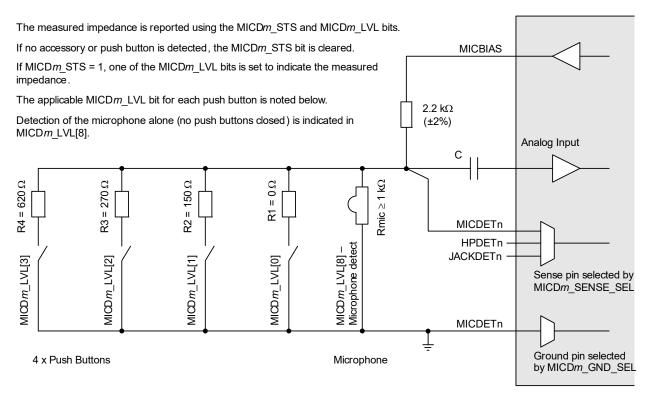


Figure 5-8. External Accessory Detect Components

5.2 Resets Summary

 Table 5-3 summarizes the CS42L92 registers and other programmable memory under different reset conditions. The associated events and conditions are listed as follows:

- A power-on reset occurs when AVDD or DBVDD is below its respective reset threshold. Note that DCVDD is also
 required for initial start-up; subsequent interruption to DCVDD should only be permitted as part of a control
 sequence for entering Sleep Mode.
- A hardware reset occurs when the RESET input is asserted (Logic 0).
- A software reset occurs when register R0 is written to.
- Sleep Mode is selected when DCVDD is removed. Note that the AVDD and DBVDD supplies must be present throughout the Sleep Mode duration.

Reset Type	Always-On Registers ¹	Other Registers	Control-Write Sequencer Memory	DSP Firmware Memory
Power-on reset	Reset	Reset	Reset	Reset
Hardware reset	Reset	Reset	Reset	Reset
Software reset	Reset	Reset	Retained	Retained ²
Sleep Mode	Retained	Reset	Reset	Reset

Table 5-3.	Memory	Reset	Summary
------------	--------	-------	---------

 $1.See\ Section\ 4.13$ for details of Sleep Mode and the always-on registers.

2. To retain the DSP firmware memory contents during software reset, it must be ensured that DCVDD is held above its reset threshold.

5.3 Output-Signal Drive-Strength Control

The CS42L92 supports configurable drive-strength control for the digital output pins. This can be used to assist system-level integration and design considerations.

The drive-strength control bits are described in Table 5-4. Note that, in the case of bidirectional pins (e.g., GPIO*n*), the drive-strength control bits are only applicable if the pin is configured as an output.



Register Address	Bit	Label	Default	Description
R8 (0x0008)	8	CIF1MISO_DRV_	1	CIF1MISO output drive strength
Ctrl IF CFG 1		STR		0 = 4 mA
				1 = 8 mA
R1520 (0x05F0)	1	SLIMDAT2_DRV_	1	SLIMDAT2 output drive strength
Slimbus_Pad_Ctrl		STR		0 = 8 mA
				1 = 12 mA
	1	SLIMDAT1_DRV_	1	SLIMDAT1 output drive strength
		STR		0 = 8 mA
				1 = 12 mA
	0	SLIMCLK_DRV_	1	SLIMCLK output drive strength
		STR		0 = 2 mA
				1 = 4 mA
R5889 (0x1701)	12	GP1_DRV_STR	1	GPIO1 output drive strength
GPIO1_CTRL2			-	0 = 4 mA
				1 = 8 mA
R5891 (0x1703)	12	GP2 DRV STR	1	GPIO2 output drive strength
GPIO2 CTRL2		····		0 = 4 mA
				1 = 8 mA
R5893 (0x1705)	12	GP3 DRV STR	1	SPKCLK/GPIO3 output drive strength
GPIO3 CTRL2				0 = 4 mA
				1 = 8 mA
R5895 (0x1707)	12	GP4 DRV STR	1	SPKDAT/GPIO4 output drive strength
GPIO4 CTRL2	12			0 = 4 mA
				1 = 8 mA
R5897 (0x1709)	12	GP5 DRV STR	1	AIF1TXDAT/GPI05 output drive strength
GPIO5_CTRL2	12	GF5_DIV_SIN	1	0 = 4 mA
				1 = 8 mA
R5899 (0x170B)	12	GP6 DRV STR	1	AIF1BCLK/GPI06 output drive strength
GPIO6_CTRL2	12	GF0_DIXV_STIX	1	0 = 4 mA
				1 = 8 mA
R5901 (0x170D)	12	GP7 DRV STR	1	AIF1RXDAT/GPIO7 output drive strength
GPIO7_CTRL2	12		1	0 = 4 mA
				1 = 8 mA
R5903 (0x170F)	12	GP8 DRV STR	1	AIF1LRCLK/GPI08 output drive strength
GPIO8_CTRL2	12	GF0_DIV_SIN	1	0 = 4 mA
				1 = 8 mA
R5905 (0x1711)	12	GP9 DRV STR	1	AIF2TXDAT/GPIO9 output drive strength
GPIO9_CTRL2	12		1	0 = 4 mA
01103_01112				1 = 8 mA
R5907 (0x1713)	12	GP10_DRV_STR	1	AIF2BCLK/GPI010 output drive strength
GPIO10_CTRL2	12		1	0 = 4 mA
				1 = 8 mA
R5909 (0x1715)	12	GP11 DRV STR	1	AIF2RXDAT/GPIO11 output drive strength
GPIO11_CTRL2	12			0 = 4 mA
				1 = 8 mA
R5911 (0x1717)	12	GP12_DRV_STR	1	AIF2LRCLK/GPI012 output drive strength
GPIO12_CTRL2	١Z		1	0 = 4 mA
GPIUIZ_UIRLZ				
BE012 (0:4740)	40		4	1 = 8 mA
R5913 (0x1719)	12	GP13_DRV_STR	1	AIF3TXDAT/GPIO13 output drive strength
GPIO13_CTRL2				0 = 4 mA
	. <u> </u>			1 = 8 mA

Table 5-4. Output Drive-Strength and Slew-Rate Control



Register Address	Bit	Label	Default	Description
R5915 (0x171B)	12	GP14_DRV_STR	1	AIF3BCLK/GPIO14 output drive strength
GPIO14_CTRL2				0 = 4 mA
				1 = 8 mA
R5917 (0x171D)	12	GP15_DRV_STR	1	AIF3RXDAT/GPIO15 output drive strength
GPIO15_CTRL2				0 = 4 mA
				1 = 8 mA
R5919 (0x171F)	12	GP16_DRV_STR	1	AIF3LRCLK/GPI016 output drive strength
GPIO16_CTRL2				0 = 4 mA
				1 = 8 mA

Table 5-4. Output Drive-Strength and Slew-Rate Control (Cont.)

5.4 Digital Audio Interface Clocking Configurations

The digital audio interfaces (AIF1–AIF3) can be configured in master or slave modes. In all applications, it is important that the system clocking configuration is correctly designed. Incorrect clock configurations lead to audible clicks arising from dropped or repeated audio samples; this is caused by the inherent tolerances of multiple asynchronous system clocks.

To ensure reliable clocking of the audio interface functions, the external interface clocks (e.g., BCLK, LRCLK) must be derived from the same clock source as SYSCLK (or ASYNCCLK, where applicable).

In AIF Master Mode, the external BCLK and LRCLK signals are generated by the CS42L92 and synchronization of these signals with SYSCLK (or ASYNCCLK) is ensured. In this case, clocking of the AIF is typically derived from the MCLK*n* inputs, either directly or via one of the FLL circuits. Alternatively, an AIF*n* or SLIMbus interface can be used to provide the reference clock to which the AIF master can be synchronized.

In AIF Slave Mode, the external BCLK and LRCLK signals are generated by another device, as inputs to the CS42L92. In this case, the system clock (SYSCLK or ASYNCCLK) must be generated from a source that is synchronized to the external BCLK and LRCLK inputs.

In a typical Slave Mode application, the BCLK input is selected as the clock reference, using the FLL to perform frequency shifting. The MCLK1, MCLK2, or MCLK3 inputs can also be used, but only if the selected clock is synchronized externally to the BCLK and LRCLK inputs. The SLIMbus interface can also provide the clock reference, via one of the FLLs, provided that the BCLK and LRCLK signals are externally synchronized with the SLIMCLK input.

The valid AIF clocking configurations are listed in Table 5-5 for AIF Master and AIF Slave Modes.

The applicable system clock (SYSCLK or ASYNCCLK) depends on the AIFn_RATE setting for the relevant digital audio interface; if AIFn_RATE < 1000, SYSCLK is applicable; if AIFn_RATE \ge 1000, ASYNCCLK is applicable.

AIF Mode	Clocking Configuration
AIF Master Mode	SYSCLK_SRC (ASYNCCLK_SRC) selects MCLK1, MCLK2, or MCLK3 as SYSCLK (ASYNCCLK) source.
	SYSCLK_SRC (ASYNCCLK_SRC) selects FLL <i>n</i> as SYSCLK (ASYNCCLK) source; FLL <i>n</i> _REFCLK_SRC selects MCLK1, MCLK2, or MCLK3 as FLL <i>n</i> source.
	SYSCLK_SRC (ASYNCCLK_SRC) selects FLL <i>n</i> as SYSCLK (ASYNCCLK) source; FLL <i>n</i> _REFCLK_SRC selects a different interface (BCLK, LRCLK, SLIMCLK) as FLL <i>n</i> source.
AIF Slave Mode	SYSCLK_SRC (ASYNCCLK_SRC) selects FLL <i>n</i> as SYSCLK (ASYNCCLK) source; FLL <i>n</i> _REFCLK_SRC selects BCLK as FLL <i>n</i> source.
	SYSCLK_SRC (ASYNCCLK_SRC) selects MCLK1, MCLK2, or MCLK3 as SYSCLK (ASYNCCLK) source, provided MCLK is externally synchronized to the BCLK input.
	SYSCLK_SRC (ASYNCCLK_SRC) selects FLL <i>n</i> as SYSCLK (ASYNCCLK) source; FLL <i>n</i> _REFCLK_SRC selects MCLK1, MCLK2, or MCLK3 as FLL <i>n</i> source, provided MCLK is externally synchronized to the BCLK input.
	SYSCLK_SRC (ASYNCCLK_SRC) selects FLL <i>n</i> as SYSCLK (ASYNCCLK) source; FLL <i>n</i> _REFCLK_SRC selects a different interface (e.g., SLIMCLK) as FLL <i>n</i> source, provided the other interface is externally synchronized to the BCLK input.

Table 5-5. AIF Clocking Configurations

In each case, the SYSCLK (ASYNCCLK) frequency must be a valid ratio to the LRCLK frequency; the supported clocking rates are defined by the SYSCLK_FREQ (ASYNC_CLK_FREQ) and SAMPLE_RATE_*n* (ASYNC_SAMPLE_RATE_*n*) fields.



The valid AIF clocking configurations are shown in Fig. 5-9 to Fig. 5-15. Note that, where MCLK1 is shown as the clock source, it is equally possible to select MCLK2 or MCLK3 as the clock source. Similarly, in cases where FLL1 is shown, it is equally possible to select FLL2.

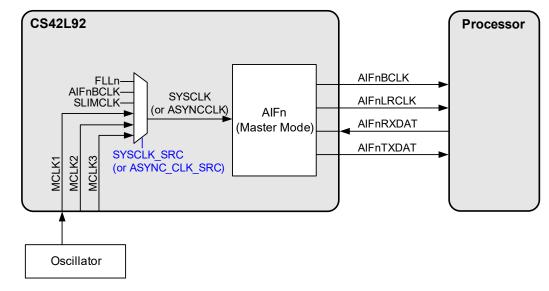


Fig. 5-9 shows AIF Master Mode operation, using MCLK as the clock reference.

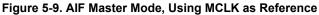


Fig. 5-10 shows AIF Master Mode operation, using MCLK as the clock reference. In this example, the FLL is used to generate the system clock, with MCLK as the reference.

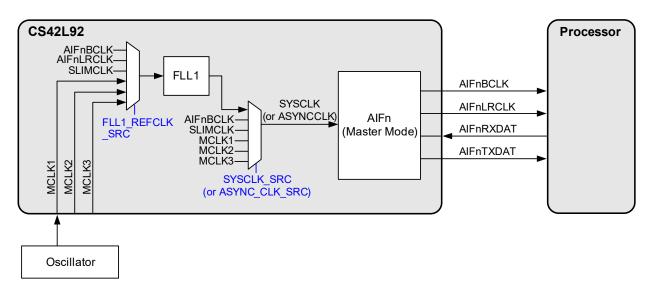


Figure 5-10. AIF Master Mode, Using MCLK and FLL as Reference



Fig. 5-11 shows AIF Master Mode operation, using a separate interface as the clock reference. In this example, the FLL is used to generate the system clock, with SLIMCLK as the reference.

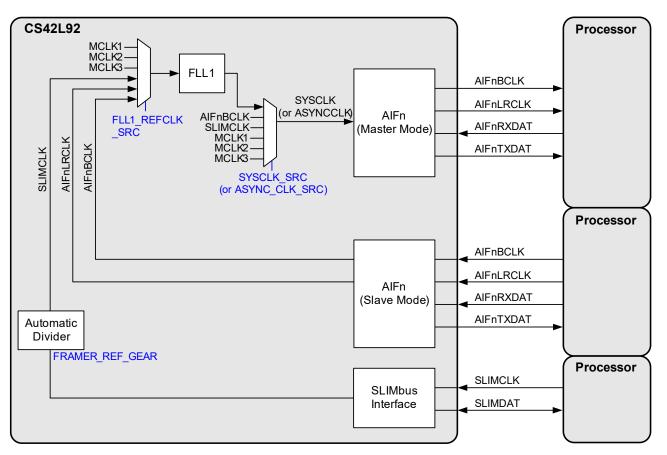


Figure 5-11. AIF Master Mode, Using Another Interface as Reference

Fig. 5-12 shows AIF Slave Mode operation, using BCLK as the clock reference. In this example, the FLL is used to generate the system clock, with BCLK as the reference.

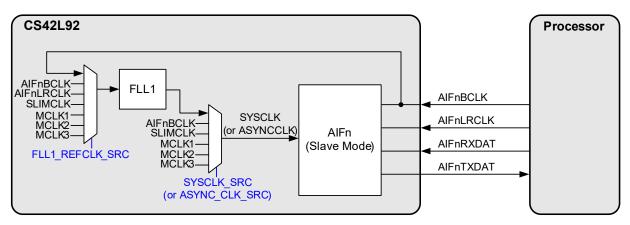


Figure 5-12. AIF Slave Mode, Using BCLK and FLL as Reference



Fig. 5-13 shows AIF Slave Mode operation, using MCLK as the clock reference. For correct operation, the MCLK input must be fully synchronized to the audio interface.

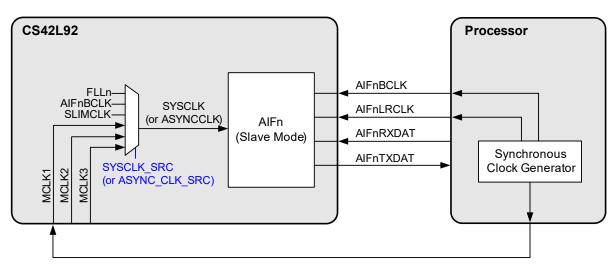


Figure 5-13. AIF Slave Mode, Using MCLK as Reference

Fig. 5-14 shows AIF Slave Mode operation, using MCLK as the clock reference. For correct operation, the MCLK input must be fully synchronized to the audio interface. In this example, the FLL is used to generate the system clock, with MCLK as the reference.

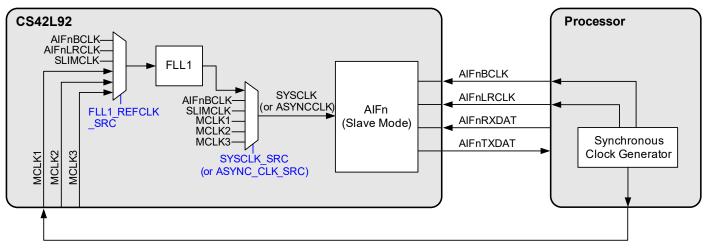


Figure 5-14. AIF Slave Mode, Using MCLK and FLL as Reference



Fig. 5-15 shows AIF Slave Mode operation, using a separate interface as the clock reference. In this example, the FLL is used to generate the system clock, with SLIMCLK as the reference. For correct operation, the SLIMCLK input must be fully synchronized to the other audio interfaces.

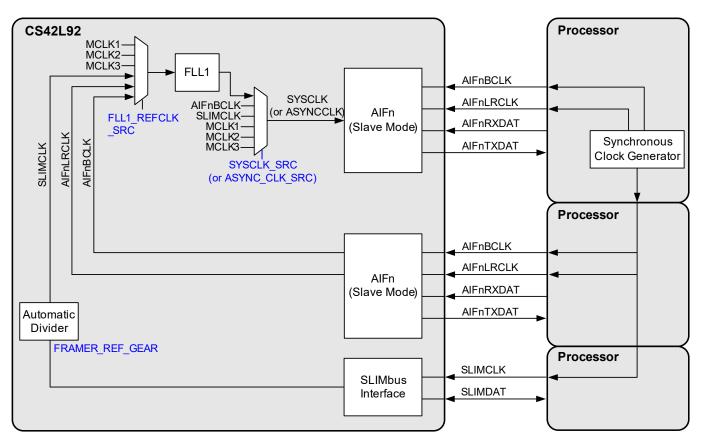


Figure 5-15. AIF Slave Mode, Using Another Interface as Reference

5.5 PCB Layout Considerations

PCB layout should be carefully considered, to ensure optimum performance of the CS42L92. Poor PCB layout degrades the performance and is a contributory factor in EMI, ground bounce, and resistive voltage losses. All external components should be placed close to the CS42L92, with current loop areas kept as small as possible. The following specific considerations should be noted:

- Placement of the charge pump capacitors is a high priority requirement—these capacitors (particularly the fly-back capacitors) must be placed as close as possible to the CS42L92. The component choice and positioning of the CP1 components are more critical than those of CP2, due to the higher output power requirements of CP1.
- Decoupling capacitors should be placed as close as possible to the CS42L92. The connection between AGND, the AVDD decoupling capacitor, and the main system ground should be made at a single point as close as possible to the AGND balls of the CS42L92.
- The VREFC capacitor should be placed as close as possible to the CS42L92. The ground connection to the VREFC capacitor should be as close as possible to the AGND1 ball of the CS42L92.
- If multiple power supply rails are connected to a single supply source, it is recommended to provide separate PCB tracks connecting each rail to the supply. This configuration is also known as *star connection*.
- If power supply rails are routed between different layers of the PCB, it is recommended to use several track vias, in
 order to minimize resistive voltage losses.
- Differential input signal tracks should be routed as a pair, ensuring similar length/width dimensions on each track. Input signal paths should be kept away from high frequency digital signals.



- Differential output signal tracks should be routed as a pair, ensuring similar length/width dimensions on each track. The tracks should provide a low resistance path from the device output pin to the load (< 1% of the minimum load).
- The headphone output ground-feedback pins should be connected to GND as close as possible to the respective headphone jack ground pin. The ground-feedback PCB track should follow the same route as the respective output signal paths.

6 Register Map

The CS42L92 control registers are listed in the following tables. Note that only the register addresses described here should be accessed; writing to other addresses may result in undefined behavior. Register bits that are not documented should not be changed from the default values.

The CS42L92 register map is defined in two regions:

- The register space below 0x3000 is defined in 16-bit word format
- The register space from 0x3000 upwards is defined in 32-bit word format

It is important to ensure that all control interface register operations use the applicable data word format, in accordance with the applicable register addresses.

The 16-bit codec register space is described in Table 6-1.

Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R0 (0h)	Software_Reset							5	SW_RST_D	EV_ID [15:0	0]							6371h
R1 (1h)	Hardware_Revision	0	0	0	0	0	0	0	0 HW_REVISION [7:0]									
R2 (2h)	Software_Revision	0	0	0	0	0	0	0	0				SW_REV	ISION [7:0]				0000h
R8 (8h)	Ctrl_IF_CFG_1	0	0	0	0	0	0	1	CIF1MISO _DRV_ _STR	CIF1MISO _PD	0	0	0	1	0	0	0	0308h
R22 (16h)	Write_Sequencer_Ctrl_0	0	0	0	0	WSEQ_ ABORT	WSEQ_ START	WSEQ_ ENA				WSEQ_	START_IN	DEX [8:0]				0000h
R23 (17h)	Write_Sequencer_Ctrl_1	0	0	0	0	0	0	WSEQ_ BUSY				WSEQ_C	URRENT_I	NDEX [8:0]				0000h
R24 (18h)	Write_Sequencer_Ctrl_2	0	0	0	0	0	0	0	0	0	0	0	0	0	0	WSEQ_ BOOT_ START	WSEQ_ LOAD_ MEM	0000h
R32 (20h)	Tone_Generator_1		ТО	NE_RATE [4:0]		0	TONE_OF	FSET [1:0]	0	0	TONE2_ OVD	TONE1_ OVD	0	0	TONE2_ ENA	TONE1_ ENA	0000h
R33 (21h)	Tone_Generator_2						•		TONE1_I	LVL [23:8]	•		•		•			1000h
R34 (22h)	Tone_Generator_3	0	0	0	0	0	0	0	0 TONE1_LVL [7:0]									0000h
R35 (23h)	Tone_Generator_4								TONE2_LVL [23:8]									1000h
R36 (24h)	Tone_Generator_5	0	0	0	0	0	0	0	0				TONE2	_LVL [7:0]				0000h
R48 (30h)	PWM_Drive_1		PV	VM_RATE [4	4:0]		PWN	I_CLK_SEL	[2:0]	0	0	PWM2_ OVD	PWM1_ OVD	0	0	PWM2_ ENA	PWM1_ ENA	0000h
R49 (31h)	PWM_Drive_2	0	0	0	0	0	0					PWM1_	LVL [9:0]					0100h
R50 (32h)	PWM_Drive_3	0	0	0	0	0	0					PWM2_	LVL [9:0]					0100h
R66 (42h)R65 (41h)	Spare_Triggers	WSEQ_ TRG16	WSEQ_ TRG15	WSEQ_ TRG14	WSEQ_ TRG13	WSEQ_ TRG12	WSEQ_ TRG11	WSEQ_ TRG10	WSEQ_ TRG9	WSEQ_ TRG8	WSEQ_ TRG7	WSEQ_ TRG6	WSEQ_ TRG5	WSEQ_ TRG4	WSEQ_ TRG3	WSEQ_ TRG2	WSEQ_ TRG1	0000h
R75 (4Bh)	Spare_Sequence_ Select_1	0	0	0	0	0	0	0	WSEQ_TRG1_INDEX [8:0]								01FFh	
R76 (4Ch)	Spare_Sequence_ Select_2	0	0	0	0	0	0	0	WSEQ_TRG2_INDEX [8:0]								01FFh	
R77 (4Dh)	Spare_Sequence_ Select_3	0	0	0	0	0	0	0	WSEQ_TRG3_INDEX [8:0]								01FFh	
R78 (4Eh)	 Spare_Sequence Select_4	0	0	0	0	0	0	0	WSEQ_TRG4_INDEX [8:0]								01FFh	
R79 (4Fh)	 Spare_Sequence Select_5	0	0	0	0	0	0	0	WSEQ_TRG5_INDEX [8:0]								01FFh	
R80 (50h)		0	0	0	0	0	0	0				WSEQ_	_TRG6_INE	DEX [8:0]				01FFh

Table 6-1. Register Map Definition—16-bit region

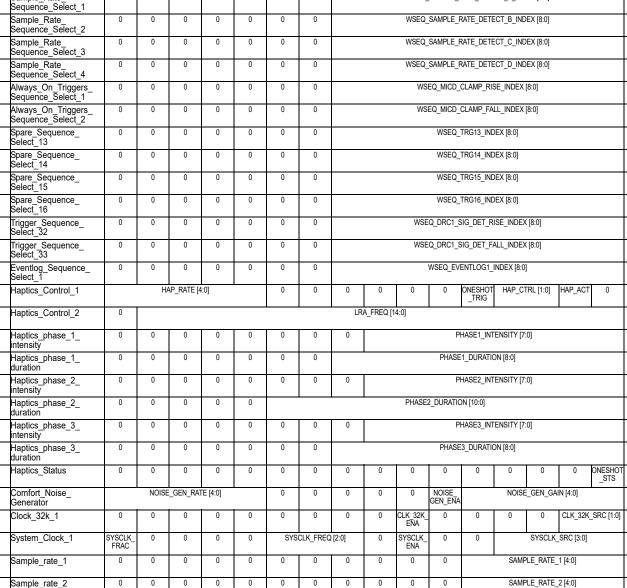


Table 6-1. Register Map Definition—16-bit region (Cont.) 9

0

0

0

0

0

0

0

8

7

6

5

4

WSEQ_TRG7_INDEX [8:0]

WSEQ_TRG8_INDEX [8:0]

WSEQ_TRG9_INDEX [8:0]

WSEQ TRG10 INDEX [8:0]

WSEQ_TRG11_INDEX [8:0]

WSEQ_TRG12_INDEX [8:0]

WSEQ_SAMPLE_RATE_DETECT_A_INDEX [8:0]

3

2

1

Name

Spare Sequence

Spare_Sequence_

Spare_Sequence_ elect 9

Spare_Sequence_

Spare_Sequence_ Select_11

Spare Sequence

elect_7

elect 8

elect_10

elect_12

Sample Rate

15

0

0

0

0

0

0

0

14

0

0

0

0

0

0

0

13

0

0

0

0

0

0

0

12

0

0

0

0

0

0

0

11

0

0

0

0

0

0

10

0

0

0

0

0

0

0

Register

R89

(59h)

R90

(5Ah)

R91

(5Bh)

R92

(5Ch)

R93

(5Dh) R94

(5Eh)

R97

(61h) R98

(62h) R99

(63h)

R100

(64h)

R102

(66h)

R103

(67h) R104

(68h)

R105

(69h)

R106

(6Ah)

R107

(6Bh)

R110

(6Eh)

R111 (6Fh)

R120

(78h) R144

(90h) R145

(91h) R146

(92h)

R147

(93h)

R148

(94h)

R149

(95h)

R150

(96h)

R151

(97h)

R152

(98h)

R160

(A0h)

R256

(100h)

R257

(101h) R258

(102h) R259

(103h)

R260

(104h) R266

(10Ah)

R267

(10Bh)

R268

(10Ch)

R274

(112h)

Sample_rate_3

Sample_rate_1_status

Sample_rate_2_status

Sample_rate_3_status

Async_clock_1

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

ASYNC_CLK_FREQ [2:0]

0

0

0

0

0

0

0

0

0

0

0

0

0

ASYNC

0

0

0

0

0

0

Default

01FFh

0000h

7FFFh

0000h

0000h

0000h

0000h

0000h

0000h

0000h

0000h

0002h

0404h

0011h

0011h

0011h

0000h

0000h

0000h

0305h

SAMPLE_RATE_3 [4:0]

SAMPLE_RATE_1_STS [4:0]

SAMPLE_RATE_2_STS [4:0]

SAMPLE_RATE_3_STS [4:0]

ASYNC_CLK_SRC [3:0]

0

	Table 6-1. Register Map Definition—16-bit region (Cont.)																	
Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R275 (113h)	Async_sample_rate_1	0	0	0	0	0	0	0	0	0	0	0		ASYNC_S	SAMPLE_R	ATE_1 [4:0]		0011h
R276 (114h)	Async_sample_rate_2	0	0	0	0	0	0	0	0									
R283 (11Bh)	Async_sample_rate_1_ status	0	0	0	0	0	0	0	0	0 0 0 ASYNC_SAMPLE_RATE_1_STS [4:0]								
R284 (11Ch)	Async_sample_rate_2_ status	0	0	0	0	0	0	0	0	0	0	0		SYNC_SAM	MPLE_RATE			0000h
R329 (149h)	Output_system_clock	OPCLK_ ENA	0	0	0	0	0	0	0			CLK_DIV [-			CLK_SEL		0000h
R330 (14Ah)	Output_async_clock	OPCLK_ ASYNC_ ENA	0	0	0	0	0	0	0		OPCLK	_ASYNC_[DIV [4:0]		OPCLK	_ASYNC_9	SEL [2:0]	0000h
R334 (14Eh)	Clock_Gen_Pad_Ctrl	0	0	0	0	0	0	MCLK3_ PD	MCLK2_ PD	MCLK1_ PD	0	0	0	0	0	0	0	0000h
R338 (152h)	Rate_Estimator_1	0	0	0	0	0	0	0	0	0	0	0	trig on Startup		CLK_SRC	-	RATE EST_EÑA	0000h
R339 (153h)	Rate_Estimator_2	0	0	0	0	0	0	0	0	0	0	0			RATE_DET		-	0000h
R340 (154h)	Rate_Estimator_3	0	0	0	0	0	0	0	0	0	0	0		-	RATE_DET		-	0000h
R341 (155h)	Rate_Estimator_4	0	0	0	0	0	0	0	0	0	0	0		-	RATE_DET		-	0000h
R342 (156h)	Rate_Estimator_5	0	0	0	0	0	0	0	0	0	0	0		-	RATE_DET		-	0000h
R352 (160h)	Clocking_debug_5	_	CLK_FREG	,			SRC_STS		0	0		K_FREQ_S			SYSCLK_SI		-	0000h
R369 (171h)	FLL1_Control_1		-	LK_SRC [3:	-	0	0	0	0	0	0	0	0	0	fll1 Hold	0	FLL1_ENA	7004h
R370 (172h)	FLL1_Control_2	FLL1_ CTRL_ UPD	0	0	0	0	0					FLL1_	N [9:0]					0004h
R371 (173h)	FLL1_Control_3								FLL1_TH	ETA [15:0]								0000h
R372 (174h)	FLL1_Control_4						FLL1_LAMBDA [15:0]										0000h	
R373 (175h)	FLL1_Control_5	0	0	0	0	0	0						_DIV [9:0]					0001h
R374 (176h)	FLL1_Control_6	FLL1 REFDET	0	0	0	0	0	0	0	[1	FCLK_DIV :0]	0	0	0	0	0	0	8000h
R376 (178h)	FLL1_Control_8			AIN_FINE [3	-		1_PD_GAIN				.L1_FD_GA		-		1_FD_GAIN			21F0h
R378 (17Ah)	FLL1_Control_10		HP [1:0]	0	1	0	0	0	0	0	0	0	0	0	0	0	0	1000h
R379 (17Bh)	FLL1_Control_11	0	0	0	0	0	0	0	0	0	0	0		_	DET_THR [3	-	FLL1 LOCKDET	0011h
R381 (17Dh)	FLL1_Digital_Test_1	0	0	1	1	0	0	1	1	0	1	1	1	1	0	FLL1_C	LK_VCO_ SRC[1:0]	33E8h
R398 (18Eh)	FLL1_GPIO_Clock	0	0	0	0	FLL1_GP	PDIV_SRC :0]	0	0			FLL1	_GPCLK_D	IV [6:0]			FLL1 GPCLK ENA	0C04h
(191h)	FLL2_Control_1		_	LK_SRC [3:	-	0	0	0	0	0	0	0	0	0	FLL2 HOLD	0	FLL2_ENA	
R402 (192h)	FLL2_Control_2	FLL2_ CTRL_ UPD	0	0	0	0	0					FLL2_	N [9:0]					0004h
R403 (193h)	FLL2_Control_3		1						FLL2_TH	ETA [15:0]								0000h
R404 (194h)	FLL2_Control_4							FLL2_LAMBDA [15:0]								0000h		
R405 (195h)	FLL2_Control_5	0	0	0	0	0	0					FLL2_FB	_DIV [9:0]					0001h
R406 (196h)	FLL2_Control_6	FLL2 REFDET	0	0	0	0	0	0	0	- [1	FCLK_DIV :0]	0	0	0	0	0	0	8000h
R408 (198h)	FLL2_Control_8			AIN_FINE [3	:0]		2_PD_GAIN	AIN_COARSE [3:0] FLL2_FD_GAIN_FINE [3:0] FLL2_FD_GAIN_COARSE [3:0]								21F0h		
R410 (19Ah)	FLL2_Control_10		HP [1:0]	0	1	0	0	0	0	0	0	0	0	0	0	0	0	1000h
R411 (19Bh)	FLL2_Control_11	0	0	0	0	0	0	0	0	0	0	0		LL2_LOCKE	Det_thr (3		FLL2_ LOCKDET	0011h
R413 (19Dh)	FLL2_Digital_Test_1	0	0	1	1	0	0	1	1	0	1	1	1	1	0	FLL2_C FAST_S	LK_VCO_ SRC[1:0]	33E8h
R430 (1AEh)	FLL2_GPIO_Clock	0	0	0	0	FLL2_GP [1	PDIV_SRC :0]	0	0			FLL2_	_GPCLK_D	IV [6:0]			FLL2 GPCLK_ ENA	0C04h
R512 (200h)	Mic_Charge_Pump_1	0	0	0	0	0	0	0	0	0	0	0	0	0	CP2_ DISCH	CP2 BYPASS	CP2_ENA	0006h
R531 (213h)	LDO2_Control_1	0	0	0	0	0			LDO2_V	SEL [5:0]			0	0	LDO2 DISCH	0	0	03E4h
·	•		•										•					

Table 6-1. Register Map Definition—16-bit region (Cont.)

CIRRUS LOGIC[®]



Table 6-1.	Register Map Definition—16-bit region (Cont.)
------------	---

RAMP MARTINATION C MARTINATION C MARTINATION C MARTINATION MARTINATION <th< th=""><th>Register</th><th>Name</th><th>15</th><th>14</th><th>13</th><th>12</th><th>11</th><th>10</th><th>9</th><th>8</th><th>7</th><th>6</th><th>5</th><th>4</th><th>3</th><th>2</th><th>1</th><th>0</th><th>Default</th></th<>	Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
1 m m< m< <td></td> <td>Mic_Bias_Ctrl_1</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td></td> <td>MICB1_</td> <td>LVL [3:0]</td> <td>•</td> <td>0</td> <td></td> <td></td> <td></td> <td></td> <td>00E6h</td>		Mic_Bias_Ctrl_1		0	0	0	0	0	0		MICB1_	LVL [3:0]	•	0					00E6h
Crich - - - - - - -	R537	Mic_Bias_Ctrl_2	MICB2_ EXT_CAP	0	0	0	0	0	0		MICB2_	LVL [3:0]		0					00E6h
C1CE - - - - - - DOCH DAX DOCH DAX DBS Maciphone_Detect_D M*D_M		Mic_Bias_Ctrl_5	0	0			0	0			0	0			0	0	MICB1A_ DISCH	MICB1A_ ENA	2222h
Clamp O <td></td> <td>Mic_Bias_Ctrl_6</td> <td>0</td> <td></td> <td></td> <td>0</td> <td>0</td> <td></td> <td></td> <td>0022h</td>		Mic_Bias_Ctrl_6	0	0	0	0	0	0	0	0	0	0			0	0			0022h
(pate) (pate)<		Headphone_Detect_0		HPD	_OUT_SEL	[2:0]		HPD_FRC	_SEL [3:0]			HPD_SENS	E_SEL [3:0]	0	HPD	_GND_SEL	. [2:0]	0000h
Cyach O <td></td> <td>Headphone_Detect_1</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td></td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>HPD_CLK</td> <td>(_DIV [1:0]</td> <td>HPD_R/</td> <td>ATE [1:0]</td> <td></td> <td>0000h</td>		Headphone_Detect_1	0	0	0	0	0			0	0	0	0	HPD_CLK	(_DIV [1:0]	HPD_R/	ATE [1:0]		0000h
(2000) - <td></td> <td>Headphone_Detect_2</td> <td>HPD DONE</td> <td></td> <td></td> <td></td> <td></td> <td></td> <td></td> <td>Н</td> <td>PD_LVL [14</td> <td>:0]</td> <td></td> <td></td> <td></td> <td></td> <td></td> <td></td> <td>0000h</td>		Headphone_Detect_2	HPD DONE							Н	PD_LVL [14	:0]							0000h
		Headphone_Detect_3	0	0	0	0	0	0					HPD_DA	CVAL [9:0]					0000h
APPTG Mic Debect 1_Control_1 MICD1_BAA_STATTIME [3:0] MICD1_MATE [3:0] MICD1_BAA_STRC [3:1] 0 0 MICD1_MATE [3:1] 0 0 0 0 MICD1_MATE [3:1] MICD1_MATE [3:1] 0 0 MICD1_MATE [3:1] 0 0 0 0 MICD1_MATE [3:1] MICD1_MATE [3:1] 0 0 MICD1_MATE [3:1] 0 0 MICD1_MATE [3:1] 0 0 0 0 0 0 0 MICD1_MATE [3:1] MICD1_MATE [3:1] MICD1_MATE [3:1] MICD1_MATE [3:1] 0 MICD1_MATE [3:1] MICD1_MATE [3:1] MICD1_MATE [3:1] MICD1_MATE [3:1] 0 MICD1_MATE [3:1] MI		Mic_Detect_1_Control_0	ADC	0	0	0	0	0	0	0	N	IICD1_SEN	SE_SEL [3:	0]	0	MICD	1_GND_SE	L [2:0]	0010h
APTG Mic_Detect_1_Control_2 0 0 0 0 0 0 MICD_UAL_SEL [74] 000Fh R877 Mic_Detect_1_Control_3 0 0 0 0 0 MICD_UAL_SEL [74] 000Fh R877 Mic_Detect_1_Control_4 MICD_UACRA_DIFF [74] 0 MICD_UACRA_E [66] 00000h R880 Mic_Detect_2_Control_4 MICD_UACRA_UFF [74] 0 MICD_UAR_SEC [76] 0		Mic_Detect_1_Control_1		D1_BIAS_S	TARTTIME	[3:0]		MICD1_F	ATE [3:0]		1	MICD1_BIA	S_SRC [3:0]	0	0			1102h
Terr Mic_Delect_1_Control_3 0 <td>R676</td> <td>Mic_Detect_1_Control_2</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td></td> <td></td> <td></td> <td>MICD1_LVI</td> <td>L_SEL [7:0]</td> <td> </td> <td></td> <td>1</td> <td>009Fh</td>	R676	Mic_Detect_1_Control_2	0	0	0	0	0	0	0	0				MICD1_LVI	L_SEL [7:0]			1	009Fh
Resc (A2Ab) Mic_Detect_1_Control_4 MCD1_ACXAL_DPF(76) 0 MCD1_ACXAL_DP(75) 0 MCD1_ACXAL_DP(75) 0 0 MCD1_ACXAL_DP(75) 0	R677	Mic_Detect_1_Control_3	0	0	0	0	0				MI	CD1_LVL [8	3:0]					MICD1_ STS	0000h
PHB00 Mc_Detect_2_Control_0 MCD2_MADE 0	R683	Mic_Detect_1_Control_4			MI	CD1_ADC\	AL_DIFF [7	7:0]			0			MICE	D1_ADCVAL	_ [6:0]	ı		0000h
RRF1 Mic_Detect_2_control_1 MICD2_BARS_SRC [5:0] 0 <td>R690</td> <td>Mic_Detect_2_Control_0</td> <td>MICD2_ ADC_ MODF</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>N</td> <td>IICD2_SEN</td> <td>SE_SEL [3:</td> <td>0]</td> <td>0</td> <td>MICD</td> <td>2_GND_SE</td> <td>L [2:0]</td> <td>0010h</td>	R690	Mic_Detect_2_Control_0	MICD2_ ADC_ MODF	0	0	0	0	0	0	0	N	IICD2_SEN	SE_SEL [3:	0]	0	MICD	2_GND_SE	L [2:0]	0010h
IPR02 (284h) Mic. Detect. 2. Control. 2 0 0 0 0 0 0 Mic. Detect. 2. Control. 3 0 0 0 0 0 0 0 0000h Mic.D2_LM_[80] Mic.D2_LM_[80] Mic.D2_LM_[80] Mic.D2_MCD2_MIC		Mic_Detect_2_Control_1		02_BIAS_S	TARTTIME	[3:0]		MICD2_F	ATE [3:0]	1	1	MICD2_BIA	S_SRC [3:0]	0	0	MICD2 DBTIME		1102h
C2B0h C <thc< th=""> C C C</thc<>	R692	Mic_Detect_2_Control_2	0	0	0	0	0	0	0	0				MICD2_LVI	L_SEL [7:0]		1		009Fh
(28Bh) C <td></td> <td>Mic_Detect_2_Control_3</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td></td> <td></td> <td>-</td> <td>MI</td> <td>CD2_LVL [8</td> <td>3:0]</td> <td></td> <td></td> <td></td> <td>MICD2_ VALID</td> <td>MICD2_ STS</td> <td>0000h</td>		Mic_Detect_2_Control_3	0	0	0	0	0			-	MI	CD2_LVL [8	3:0]				MICD2_ VALID	MICD2_ STS	0000h
(205h) Int_annum Image: Normal and the second of the sec		Mic_Detect_2_Control_4			M	CD2_ADC\	AL_DIFF [7	7:0]			0 MICD2_ADCVAL [6:0]							0000h	
C2C0B1 Call Call Call Call Call Call<		Micd_Clamp_control	0	0	0	0	0	0	CLAMP2_	MICD_C	LAMP2_MC	DDE [2:0]	0	CLAMP1_	MI	CD_CLAMF	3:0]	0210h	
(2D3h) - - O OO O </td <td></td> <td>GP_Switch_1</td> <td>0</td> <td>SW2_M0</td> <td>ODE [1:0]</td> <td>SW1_M</td> <td>ODE [1:0]</td> <td>0000h</td>		GP_Switch_1	0	0	0	0	0	0	0	0	0	0	0	0	SW2_M0	ODE [1:0]	SW1_M	ODE [1:0]	0000h
(300h) Int		Jack_detect_analogue	0		0	0	0	0	0	0	0	0	0	0	0	JD3_ENA	JD2_ENA	JD1_ENA	0000h
(301h) Image		Input_Enables	0	0	0	0	0	0	0	0	IN4L_ENA	IN4R_ENA	IN3L_ENA	IN3R_ENA	IN2L_ENA	IN2R_ENA	IN1L_ENA	IN1R_ENA	0000h
$ \begin{array}{ c c c c c c c c c c c c c c c c c c c$		Input_Enables_Status	0	0	0	0	0	0	0	0				IN3R_ ENA_STS	IN2L_ ENA_STS	IN2R_ ENA_STS	IN1L_ ENA_STS	IN1R_ ENA_STS	0000h
(309h) Image: Control interpretation of the interpretation of t		Input_Rate		11	I_RATE [4:	0]			0	0	0	0	0	0	0	0	0	0	0400h
(30Ch) - 0 <td>(309h)</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>IN_</td> <td>VD_RAMP</td> <td>[2:0]</td> <td>0</td> <td>_</td> <td></td> <td>-</td> <td>0022h</td>	(309h)		0	0	0	0	0	0	0	0	0	IN_	VD_RAMP	[2:0]	0	_		-	0022h
(310h) - - - MODE MODE - 0 <t< td=""><td>(30Ch)</td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>IN_</td><td>HPF_CUT</td><td>[2:0]</td><td>0002h</td></t<>	(30Ch)		0	0	0	0	0	0	0	0	0	0	0	0	0	IN_	HPF_CUT	[2:0]	0002h
(311h) Image: Control MIL_SIG DET_ENA 0		IN1L_Control	IN1L_HPF	0	0	IN1_DMIC	_SUP [1:0]	IN1_ MODE		0			IN1L	PGA_VOL	[6:0]			0	
R787 (313h) N1L_Rate_Control INIL_RATE [4:0] 0	(311h)			IN1L_S	RC [1:0]	0			_	MUTE				IN1L_V	'OL [7:0]				
(313h) - <td></td> <td>DMIC1L_Control</td> <td>IN1L_SIG_ DET_ENA</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>11</td> <td>1_0SR [2:</td> <td>:0]</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0500h</td>		DMIC1L_Control	IN1L_SIG_ DET_ENA	0	0	0	0	11	1_0SR [2:	:0]	0	0	0	0	0	0	0	0	0500h
$ \begin{array}{c c c c c c c c c c c c c c c c c c c $	(313h)			IN	IL_RATE [4	:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
(315h) Image: Control (315h) Image: Control (315h) Image: Control (316h) Image: Control (317h) Image: Control (317h) Image: Control (318h)		IN1R_Control	IN1R_HPF	0	0	0	0	0	0	0			IN1R	_PGA_VOL	[6:0]		-	0	0080h
R791 (317h) IN1R_Rate_Control IN1R_RATE [4:0] 0 </td <td>R789 (315h)</td> <td></td> <td></td> <td>IN1R_S</td> <td>RC [1:0]</td> <td>0</td> <td>IN1R_LP_ MODE</td> <td>0</td> <td>IN_VU</td> <td>IN1R MUTE</td> <td></td> <td></td> <td></td> <td>IN1R_V</td> <td>'OL [7:0]</td> <td></td> <td></td> <td></td> <td>0180h</td>	R789 (315h)			IN1R_S	RC [1:0]	0	IN1R_LP_ MODE	0	IN_VU	IN1R MUTE				IN1R_V	'OL [7:0]				0180h
R791 (317h) IN1R_Rate_Control IN1R_RATE [4:0] 0		DMIC1R_Control	IN1R_ SIG_DET_ ENA	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R792 (318h) IN2L_Control IN2L_HPF 0 0 IN2_DMIC_SUP [1:0] IN2 MODE 0 0 IN2L_PGA_VOL [6:0] 0 0 0080h R793 (319h) ADC_Digital_Volume_2L 0 IN2L_SRC [1:0] 0 IN2L_P 0 IN2L_W IN2L_VOL [7:0] 0 0180h R794 DMIC2L_Control N2L_SIG_0 0 0 0 0 0 0 0 0 0 0 0 0500h		IN1R_Rate_Control		IN′	R_RATE [4	:0]	1	0	0	0	0	0	0	0	0	0	0	0	0000h
(319h) MODE MUTE R794 DMIC2L_Control N2L_SIG_0 0	R792	IN2L_Control	IN2L_HPF	0	0	IN2_DMIC	_SUP [1:0]	IN2 MODE	0	0			IN2L	_PGA_VOL	[6:0]	1		0	0080h
R794 (31Ah) DMIC2L_Control IN2L_SIG DET_ENA 0	(319h)		0	IN2L_SI	RC [1:0]	0	IN2L_LP_ MODE	0	IN_VU	IN2L MUTE				IN2L_V	'OL [7:0]				0180h
	R794 (31Ah)	DMIC2L_Control	IN2L_SIG_ DET_ENA	0	0	0	0	11	12_OSR [2:	:0]	0	0	0	0	0	0	0	0	0500h

Table 6-1. Register Map Definition—16-bit region (Cont.)	
--	--

Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R795 (31Bh)	IN2L_Rate_Control		IN	2L_RATE [4	4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R796 (31Ch)	IN2R_Control	IN2R_HPF	0	0	0	0	0	0	0			IN2F	PGA_VOL	[6:0]			0	0080h
R797 (31Dh)	ADC_Digital_Volume_2R	0	IN2R_S	RC [1:0]	0	IN2R_LP_ MODE	0	IN_VU	IN2R MUTE				IN2R_V	'OL [7:0]			•	0180h
R798 (31Eh)	DMIC2R_Control	IN2R_ SIG_DET_ ENA	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R799 (31Fh)	IN2R_Rate_Control		IN2	2R_RATE [4	4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R800 (320h)	IN3L_Control	IN3L_HPF	0	0	IN3_DMIC	_SUP [1:0]	0	0	0	0	0	0	0	0	0	0	0	0000h
	ADC_Digital_Volume_3L	0	0	0	0	0	0	IN_VU	IN3L MUTE			L	IN3L_V	OL [7:0]				0180h
R802 (322h)	DMIC3L_Control	IN3L_SIG_ DET_ENA	0	0	0	0	I	N3_OSR [2:	0]	0	0	0	0	0	0	0	0	0500h
R803 (323h)	IN3L_Rate_Control		IN	3L_RATE [4	4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R804 (324h)	IN3R_Control	IN3R_HPF	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R805 (325h)	ADC_Digital_Volume_3R	0	0	0	0	0	0	IN_VU	IN3R MUTE				IN3R_V	OL [7:0]			•	0180h
R806 (326h)	DMIC3R_Control	IN3R SIG_DET_ ENA	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R807 (327h)	IN3R_Rate_Control		IN	BR_RATE [4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R808 (328h)	IN4L_Control	IN4L_HPF	0	0	IN4_DMIC	_SUP [1:0]	0	0	0	0	0	0	0	0	0	0	0	0000h
R809 (329h)	ADC_Digital_Volume_4L	0	0	0	0	0	0	IN_VU	IN4L MUTE		L		IN4L_V	OL [7:0]				0180h
R810 (32Ah)	DMIC4L_Control	IN4L_SIG_ DET_ENA	0	0	0	0	I	N4_OSR [2:	0]	0	0	0	0	0	0	0	0	0500h
R811 (32Bh)	IN4L_Rate_Control		IN	4L_RATE [4	4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R812 (32Ch)	IN4R_Control	IN4R_HPF	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R813 (32Dh)	ADC_Digital_Volume_4R	0	0	0	0	0	0	IN_VU	IN4R MUTE		L		IN4R_V	OL [7:0]				0180h
R814 (32Eh)	DMIC4R_Control	IN4R_ SIG_DET_ ENA	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R815 (32Fh)	IN4R_Rate_Control		IN4	4R_RATE [4	4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R832 (340h)	Signal_Detect_Globals	0	0	0	0	0	0	0		IN_SI	G_DET_TH	R [4:0]	•	I	N_SIG_DE	T_HOLD [3	:0]	0001h
R840 (348h)	Dig_Mic_Pad_Ctrl	0	0	0	0	0	0	0	0	0	0	0	0	DMICDAT4 _PD	DMICDAT3	DMICDAT: _PD	2DMICDAT1 _PD	0000h
R1024 (400h)	Output_Enables_1	EP_SEL	0	0	0	0	0	OUT5L_ ENA	OUT5R_ ENA	0	0	HP3L_ ENA	HP3R_ ENA	HP2L_ ENA	HP2R_ ENA	HP1L_ ENA	HP1R_ ENA	0000h
R1025 (401h)	Output_Status_1	0	0	0	0	0	0	OUT5L_ ENA_STS	OUT5R_ ENA_STS	0	0	0	0	0	0	0	0	0000h
R1030 (406h)	Raw_Output_Status_1	0	0	0	0	0	0	0	0	0	0	OUT3L ENA_STS	OUT3R_ ENA_STS	OUT2L_ ENA_STS	OUT2R_ ENA_STS	OUT1L ENA_STS	OUT1R_ ENA_STS	0000h
R1032 (408h)	Output_Rate_1		OL	JT_RATE [4	4:0]		0	0	0	0	CP_DAC_ MODE		CLK_DIV :0]	0	OUT	_CLK_SR	C [2:0]	0040h
R1033 (409h)	Output_Volume_Ramp	0	0	0	0	0	0	0	0	0	OUT	_VD_RAMF	P [2:0]	0	OUT	_VI_RAMF	P [2:0]	0022h
R1040 (410h)	Output_Path_Config_1L	0	OUT1L_ HIFI	0	OUT1_ MONO	0	0	0	0	1	0	0	0	0	0	0	0	0080h
R1041 (411h)	DAC_Digital_Volume_1L	0	0	0	0	0	0	OUT_VU	OUT1L_ MUTE				OUT1L_	VOL [7:0]				0180h
R1042 (412h)	Output_Path_Config_1	0	0	0	0	0	0	0	0	0	0	0	0	0	HP1	_GND_SEI	[2:0]	0000h
R1043 (413h)	Noise_Gate_Select_1L	0	0	0	0					0	UT1L_NGA	TE_SRC [1'	1:0]					0001h
R1044 (414h)	Output_Path_Config_1R		OUT1R_ HIFI	0	0	0	0	0	0	1	0	0	0	0	0	0	0	0080h
R1045 (415h)	DAC_Digital_Volume_1R	0	0	0	0	0	0	OUT_VU	OUT1R_ MUTE					VOL [7:0]				0180h
R1047 (417h)	Noise_Gate_Select_1R	0	0	0	0					0	JT1R_NGA	TE_SRC [1	1:0]					0002h
R1048 (418h)	Output_Path_Config_2L	0	OUT2L_ HIFI	0	OUT2 MONO	0	0	0	0	1	0	0	0	0	0	0	0	0080h
R1049 (419h)	DAC_Digital_Volume_2L		0	0	0	0	0	OUT_VU	OUT2L_ MUTE					VOL [7:0]				0180h
R1050 (41Ah)	Output_Path_Config_2	0	0	0	0	0	0	0	0	0	0	0	0	0	HP2	_GND_SEI	_ [2:0]	0002h

CS42L92
6 Register Map

Table 6-1. Register Map Definition—16-bit region (Cont.)

Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default		
R1051 (41Bh)	Noise_Gate_Select_2L	0	0	0	0					0	UT2L_NGA	TE_SRC [11	:0]	1			1	0004h		
	Output_Path_Config_2R	0	OUT2R_ HIFI	0	0	0	0	0	0	1	0	0	0	0	0	0	0	0080h		
	DAC_Digital_Volume_2R	0	0	0	0	0	0	OUT_VU	OUT2R_ MUTE				OUT2R_	VOL [7:0]			1	0180h		
R1055 (41Fh)	Noise_Gate_Select_2R	0	0	0	0	-			1	Ol	JT2R_NGA	TE_SRC [11	1:0]					0008h		
	Output_Path_Config_3L	0	OUT3L_ HIFI	0	OUT3_ MONO	0	0	0	0	1	0	0	0	0	0	0	0	0080h		
R1057 (421h)	DAC_Digital_Volume_3L	0	0	0	0	0	0 OUT_VU OUT3L OUT3L_VOL [7:0]											0180h		
R1058 (422h)	Output_Path_Config_3	0	0	0	0	0	0	0	0	0	[2:0]	0002h								
R1059 (423h)	Noise_Gate_Select_3L	0	0	0	0					0	UT3L_NGA	TE_SRC [11	:0]					0010h		
R1060 (424h)	Output_Path_Config_3R	0	OUT3R_ HIFI	0	0	0	0	0	0	1	0	0	0	0	0	0	0	0080h		
R1061 (425h)	DAC_Digital_Volume_3R	0	0	0	0	0	0	OUT_VU	OUT3R_ MUTE				OUT3R_	VOL [7:0]				0180h		
R1063 (427h)	Noise_Gate_Select_3R	0	0	0	0			-		Ol	JT3R_NGA	TE_SRC [11	1:0]					0020h		
R1072 (430h)	Output_Path_Config_5L	0	OUT5L_ HIFI	OUT5_ OSR	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h		
R1073 (431h)	DAC_Digital_Volume_5L	0	0	0	0	0	0	OUT_VU	OUT5L_ MUTE				OUT5L_	VOL [7:0]				0180h		
R1075 (433h)	Noise_Gate_Select_5L	0	0	0	0			-		0	UT5L_NGA	TE_SRC [11	:0]					0100h		
R1076 (434h)	Output_Path_Config_5R	0	OUT5R_ HIFI	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000h		
R1077 (435h)	DAC_Digital_Volume_5R	0	0	0	0	0	0	OUT_VU	OUT5R_ MUTE			0180h								
R1079 (437h)	Noise_Gate_Select_5R	0	0	0	0 OUT5R_NGATE_SRC [11:0]													0200h		
R1102 (44Eh)	Filter_Control	0	0	0	0	0	0	0	0	0	0	0	0000h							
R1104 (450h)	DAC_AEC_Control_1	0	0	0	0	0	0	0	0	0	0	AEC	C1_LOOPB	ACK_SRC	[3:0]	AEC1_ ENA_STS	AEC1_ LOOPBAC K ENA	0000h		
R1105 (451h)	DAC_AEC_Control_2	0	0	0	0	0	0	0	0	0	0	AEC2_LOOPBACK_SRC [3:0] AEC2_AEC2 ENA_STS LOOPB K EN						0000h		
R1112 (458h)	Noise_Gate_Control	0	0	0	0	0	0	0	0	0	0	NGATE_H	IOLD [1:0]	NG	GATE_THR	[2:0]	NGATE_ ENA	0000h		
R1168 (490h)	PDM_SPK1_CTRL_1	0	0	SPK1R_ MUTE	SPK1L_ MUTE	0	0	0	SPK1_ MUTE_ ENDIAN		1		SPK1_MUT	E_SEQ [7:0	0]			0069h		
R1169 (491h)	PDM_SPK1_CTRL_2	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	SPK1_ FMT	0000h		
	AIF1_BCLK_Ctrl	0	0	0	0	0	0	0	0	AIF1 BCLK_ĪNV	AIF1_ BCLK_ FRC	AIF1_ BCLK_ MSTR		AIF1_	BCLK_FRE	:Q [4:0]	II	000Ch		
R1281 (501h)	AIF1_Tx_Pin_Ctrl	0	0	0	0	0	0	0	0	0	0	AIF1TX_ DAT_TRI	0	0	0	0	0	0000h		
	AIF1_Rx_Pin_Ctrl	0	0	0	0	0	0	0	0	0	0	0	AIF1 LRCLK_ MODE	0	AIF1 LRCLK_ INV	AIF1 LRCLK_ FRC	AIF1 LRCLK_ MSTR	0000h		
R1283 (503h)	AIF1_Rate_Ctrl		All	F1_RATE [4	1:0]		0	0	0	0	AIF1_TRI	0	0	0	0	0	0	0000h		
	AIF1_Format	0	0	0	0	0	0	0	0	0	0	0	0	0	A	IF1_FMT [2	:0]	0000h		
	AIF1_Rx_BCLK_Rate	0	0	0					1	AIF	-1_BCPF [1	2:0]	1	1				0040h		
	AIF1_Frame_Ctrl_1	0	0			AIF1TX	_WL [5:0]					Α	NF1TX_SLO	DT_LEN [7:	0]			1818h		
	AIF1_Frame_Ctrl_2	0	0			AIF1RX	_WL [5:0]					A	IF1RX_SLO	OT_LEN [7:	0]			1818h		
	AIF1_Frame_Ctrl_3	0	0	0	0	0	0	0	0	0	0			AIF1TX1_	SLOT [5:0]			0000h		
	AIF1_Frame_Ctrl_4	0	0	0	0	0	0	0	0	0	0			AIF1TX2_	SLOT [5:0]			0001h		
	AIF1_Frame_Ctrl_5	0	0	0	0	0	0	0	0	0	0	0 AIF1TX3_SLOT [5:0]								
	AIF1_Frame_Ctrl_6	0	0	0	0	0	0	0	0	0	0			AIF1TX4_	SLOT [5:0]			0003h		
	AIF1_Frame_Ctrl_7	0	0	0	0	0	0	0	0	0	0			AIF1TX5_	SLOT [5:0]			0004h		
R1294 (50Eh)	AIF1_Frame_Ctrl_8	0	0	0	0	0	0	0	0	0	0			AIF1TX6_	0005h					



Table 6-1. Register Map Definition—16-b	oit region <i>(Cont.)</i>
---	---------------------------

Register	Name	15 0	14 0	13 0	12 0	11 0	10 0	9 0	8 0	7 0	6 0	5	4	AIF1TX7	2	1	0	Default
R1295 (50Fh)	AIF1_Frame_Ctrl_9	U	U	U	0	U	U	U	U	0	U		0006h					
R1296 (510h)	AIF1_Frame_Ctrl_10	0	0	0	0	0	0	0	0	0	0		0007h					
R1297 (511h)	AIF1_Frame_Ctrl_11	0	0	0	0	0	0	0	0	0	0		0000h					
R1298 (512h)	AIF1_Frame_Ctrl_12	0	0	0	0	0	0	0	0	0	0		0001h					
R1299 (513h)	AIF1_Frame_Ctrl_13	0	0	0	0	0	0	0	0	0	0		0002h					
R1300 (514h)	AIF1_Frame_Ctrl_14	0	0	0	0	0	0	0	0	0	0		0003h					
R1301 (515h)	AIF1_Frame_Ctrl_15	0	0	0	0	0	0	0	0	0	0		0004h					
R1302 (516h)	AIF1_Frame_Ctrl_16	0	0	0	0	0	0	0	0	0	0		0005h					
R1303 (517h)	AIF1_Frame_Ctrl_17	0	0	0	0	0	0	0	0	0	0			AIF1RX7_	SLOT [5:0]			0006h
R1304 (518h)	AIF1_Frame_Ctrl_18	0	0	0	0	0	0	0	0	0	0			AIF1RX8_	SLOT [5:0]			0007h
R1305 (519h)	AIF1_Tx_Enables	0	0	0	0	0	0	0	0	AIF1TX8_ ENA	AIF1TX7_ ENA	AIF1TX6_AIF1TX5_AIF1TX4_AIF1TX3_AIF1TX2_AIF1TX1_ ENA ENA ENA ENA ENA						0000h
R1306 (51Ah)	AIF1_Rx_Enables	0	0	0	0	0	0	0	0	AIF1RX8_ ENA	AIF1RX7_ ENA	AIF1RX6_ ENA	AIF1RX5_ ENA	AIF1RX4_ ENA	AIF1RX3_ ENA	AIF1RX2_ ENA	AIF1RX1_ ENA	0000h
R1344 (540h)	AIF2_BCLK_Ctrl	0	0	0	0	0	0	0	0	AIF2 BCLK_INV	AIF2_ BCLK_ FRC	AIF2AIF2_BCLK_FREQ [4:0] BCLK MSTR						000Ch
R1345 (541h)	AIF2_Tx_Pin_Ctrl	0	0	0	0	0	0	0	0	0	0	AIF2TX DAT_TRI	0	0	0	0	0	0000h
R1346 (542h)	AIF2_Rx_Pin_Ctrl	0	0	0	0	0	0	0	0	0	0	0	AIF2 LRCLK_ MODE	0	AIF2 LRCLK_ INV	AIF2 LRCLK_ FRC	AIF2 LRCLK_ MSTR	0000h
R1347 (543h)	AIF2_Rate_Ctrl		AI	F2_RATE [4	k:0]		0	0	0	0	AIF2_TRI	0	0					0000h
R1348 (544h)	AIF2_Format	0	0	0	0	0	0	0	0	0	0	0	0000h					
R1350 (546h)	AIF2_Rx_BCLK_Rate	0	0	0		1	AIF2_BCPF [12:0]											
R1351 (547h)	AIF2_Frame_Ctrl_1	0	0			AIF2TX	_WL [5:0]					٨	IF2TX_SL	OT_LEN [7:	0]			1818h
R1352 (548h)	AIF2_Frame_Ctrl_2	0	0			AIF2RX	_WL [5:0]					A	IF2RX_SL	OT_LEN [7:	0]			1818h
R1353 (549h)	AIF2_Frame_Ctrl_3	0	0	0	0	0	0	0	0	0	0			AIF2TX1_	SLOT [5:0]			0000h
R1354 (54Ah)	AIF2_Frame_Ctrl_4	0	0	0	0	0	0	0	0	0	0			AIF2TX2_	SLOT [5:0]			0001h
R1355 (54Bh)	AIF2_Frame_Ctrl_5	0	0	0	0	0	0	0	0	0	0			AIF2TX3_	SLOT [5:0]			0002h
R1356 (54Ch)	AIF2_Frame_Ctrl_6	0	0	0	0	0	0	0	0	0	0			AIF2TX4_	SLOT [5:0]			0003h
R1357 (54Dh)	AIF2_Frame_Ctrl_7	0	0	0	0	0	0	0	0	0	0			AIF2TX5_	SLOT [5:0]			0004h
R1358 (54Eh)	AIF2_Frame_Ctrl_8	0	0	0	0	0	0	0	0	0	0			AIF2TX6_	SLOT [5:0]			0005h
R1359 (54Fh)	AIF2_Frame_Ctrl_9	0	0	0	0	0	0	0	0	0	0			AIF2TX7_	SLOT [5:0]			0006h
R1360 (550h)	AIF2_Frame_Ctrl_10	0	0	0	0	0	0	0	0	0	0			AIF2TX8_	SLOT [5:0]			0007h
R1361 (551h)	AIF2_Frame_Ctrl_11	0	0	0	0	0	0	0	0	0	0			AIF2RX1_	SLOT [5:0]			0000h
R1362 (552h)	AIF2_Frame_Ctrl_12	0	0	0	0	0	0	0	0	0	0			AIF2RX2	SLOT [5:0]			0001h
R1363 (553h)	AIF2_Frame_Ctrl_13	0	0	0	0	0	0	0	0	0	0			AIF2RX3_	SLOT [5:0]			0002h
R1364 (554h)	AIF2_Frame_Ctrl_14	0	0	0	0	0	0	0	0	0	0			AIF2RX4_	SLOT [5:0]			0003h
R1365 (555h)	AIF2_Frame_Ctrl_15	0	0	0	0	0	0	0	0	0	0			AIF2RX5_	SLOT [5:0]			0004h
R1366 (556h)	AIF2_Frame_Ctrl_16	0	0	0	0	0	0	0	0	0	0		0005h					
R1367 (557h)	AIF2_Frame_Ctrl_17	0	0	0	0	0	0	0	0	0	0		0006h					
R1368 (558h)	AIF2_Frame_Ctrl_18	0	0	0	0	0	0	0	0	0	0		0007h					
R1369 (559h)	AIF2_Tx_Enables	0	0	0	0	0	0	0	0	AIF2TX8_ ENA	AIF2TX7_ ENA	AIF2TX6_ ENA	AIF2TX5_ ENA	AIF2TX4_ ENA	AIF2TX3_ ENA	AIF2TX2_ ENA	AIF2TX1_ ENA	0000h
R1370 (55Ah)	AIF2_Rx_Enables	0	0	0	0	0	0	0	0	AIF2RX8_ ENA	AIF2RX7_ ENA	AIF2RX6_ ENA	0000h					
	•	•	•			•	•		•	•	•	•		•				



Default

000Ch

0000h

0000h

0000h

0000h

0040h

1818h

1818h

0000h

0001h

0002h

0003h

0004h

0005h

0006h

0007h

0000h 0001h

0002h

0003h

0004h

0005h

0006h

0007h

0000h

0000h

0000h

0000h

0001h

0000h

0100h

0302h 0504h

0706h

0908h

0B0Ah

0D0Ch

0F0Eh

SLIMRX7_PORT_ADDR [7:0]

SLIMTX1_PORT_ADDR [7:0]

SLIMTX3_PORT_ADDR [7:0]

SLIMTX5_PORT_ADDR [7:0]

SLIMTX7_PORT_ADDR [7:0]

R1493

(5D5h) R1494 (5D6h)

R1495

(5D7h)

R1496

(5D8h) R1497

(5D9h)

SLIMbus_RX_Ports3

SLIMbus_TX_Ports0

SLIMbus_TX_Ports1

SLIMbus_TX_Ports2

SLIMbus_TX_Ports3

																6	Regis				
			т	able 6	-1. Re	egiste	r Map	Defin	ition–	–16-bi	t regio	on (Co	ont.)								
Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0				
R1408 (580h)	AIF3_BCLK_Ctrl	0	0	0	0	0	0	0	0	AIF3 BCLK_INV	AIF3_ BCLK_ FRC	AIF3_ BCLK_ MSTR		AIF3_	BCLK_FRE	Q [4:0]	<u> </u>				
R1409 (581h)	AIF3_Tx_Pin_Ctrl	0	0	0	0	0	0	0	0	0	0	AIF3TX DAT_TRI	0	0	0	0	0				
R1410 (582h)	AIF3_Rx_Pin_Ctrl	0	0	0	0	0	0	0	0	0	0	0	AIF3 LRCLK_ MODE	0	AIF3 LRCLK_ INV	AIF3 LRCLK_ FRC	AIF3 LRCLK_ MSTR				
R1411 (583h)	AIF3_Rate_Ctrl		A	IF3_RATE [4:0]		0	0	0	0	AIF3_TRI	0	0	0	0	0	0				
R1412 (584h)	AIF3_Format	0	0	0	0	0	0	0	0	0	0	0	0	0	A	NF3_FMT [2	2:0]				
R1414 (586h)	AIF3_Rx_BCLK_Rate	0	0	0					1	AIF	-3_BCPF [1	PF [12:0]									
R1415 (587h)	AIF3_Frame_Ctrl_1	0	0 0 AIF3TX_WL [5:0] AIF3TX_SLOT_LEN [7:0]																		
R1416 (588h)	AIF3_Frame_Ctrl_2	0	0			AIF3RX	(_WL [5:0]					A	AIF3RX_SL	OT_LEN [7	:0]						
R1417 (589h)	AIF3_Frame_Ctrl_3	0	0	0	0	0	0	0	0	0	0			AIF3TX1_	_SLOT [5:0]						
R1418 (58Ah)	AIF3_Frame_Ctrl_4	0	0	0	0	0	0	0	0	0	0			AIF3TX2_	_SLOT [5:0]						
R1419 (58Bh)	AIF3_Frame_Ctrl_5	0	0	0	0	0	0	0	0	0	0			AIF3TX3_	_SLOT [5:0]						
R1420 (58Ch)	AIF3_Frame_Ctrl_6	0	0	0	0	0	0	0	0	0	0	AIF3TX4_SLOT [5:0]									
R1421 (58Dh)	AIF3_Frame_Ctrl_7	0	0	0	0	0	0	0	0	0	0	1		AIF3TX5	_SLOT [5:0]						
R1422 (58Eh)	AIF3_Frame_Ctrl_8	0	0	0	0	0	0	0	0	0	0			AIF3TX6_	_SLOT [5:0]						
R1423 (58Fh)	AIF3_Frame_Ctrl_9	0	0	0	0	0	0	0	0	0	0	AIF3TX7_SLOT [5:0]									
R1424 (590h)	AIF3_Frame_Ctrl_10	0	0	0	0	0	0	0	0	0	0	AIF3TX8_SLOT [5:0]									
R1425 (591h)	AIF3_Frame_Ctrl_11	0	0	0	0	0	0	0	0	0	0			AIF3RX1_	_SLOT [5:0]						
R1426 (592h)	AIF3_Frame_Ctrl_12	0	0	0	0	0	0	0	0	0	0			AIF3RX2	_SLOT [5:0]						
R1427 (593h)	AIF3_Frame_Ctrl_13	0	0	0	0	0	0	0	0	0	0			AIF3RX3	_SLOT [5:0]						
R1428 (594h)	AIF3_Frame_Ctrl_14	0	0	0	0	0	0	0	0	0	0			AIF3RX4	_SLOT [5:0]						
R1429 (595h)	AIF3_Frame_Ctrl_15	0	0	0	0	0	0	0	0	0	0			AIF3RX5	_SLOT [5:0]						
R1430 (596h)	AIF3_Frame_Ctrl_16	0	0	0	0	0	0	0	0	0	0			AIF3RX6	_SLOT [5:0]						
R1431 (597h)	AIF3_Frame_Ctrl_17	0	0	0	0	0	0	0	0	0	0			AIF3RX7	_SLOT [5:0]						
R1432 (598h)	AIF3_Frame_Ctrl_18	0	0	0	0	0	0	0	0	0	0			AIF3RX8	_SLOT [5:0]						
R1433 (599h)	AIF3_Tx_Enables	0	0	0	0	0	0	0	0	AIF3TX8_ ENA	AIF3TX7_ ENA	AIF3TX6_ ENA	AIF3TX5_ ENA	AIF3TX4 ENA	AIF3TX3_ ENA	AIF3TX2_ ENA	_ AIF3TX1_ ENA				
R1434 (59Ah)	AIF3_Rx_Enables	0	0	0	0	0	0	0	0	AIF3RX8_ ENA	AIF3RX7_ ENA	AIF3RX6_ ENA	AIF3RX5_ ENA	AIF3RX4 ENA	AIF3RX3_ ENA	AIF3RX2_ ENA	_AIF3RX1_ ENA				
R1474 (5C2h)	SPD1_TX_Control	0	0	SPD1_ VAL2	SPD1_ VAL1	0	0	0		SP	PD1_RATE	[4:0]		0	0	0	SPD1_ ENA				
R1475 (5C3h)	SPD1_TX_Channel_ Status_1				SPD1_CAT	rcode [7:0	[0			SPD1_CHSTMODE SPD1_PREEMPH [2:0] SPD1_ S											
R1476 (5C4h)	SPD1_TX_Channel_ Status_2		SPD1_FREQ [3:0] SPD1_CHNUM2 [3:0]								SPD1_CHNUM1 [3:0] SPD1_SRCNUM [3:0]										
R1477 (5C5h)	SPD1_TX_Channel_ Status_3	0	0	0	0		SPD1_OR	GSAMP [3:0	0]	SPD1_TXWL [2:0] SPD1_ MAXWL SPD1_CS31_30 [1:0] SPD1_CLKACU [1:0]											
R1490 (5D2h)	SLIMbus_RX_Ports0			SL	IMRX2_PO	RT_ADDR	[7:0]					SL	IMRX1_PO	RT_ADDR	[7:0]						
R1491 (5D3h)	SLIMbus_RX_Ports1			SL	IMRX4_PO	RT_ADDR	[7:0]					SL	IMRX3_PO	RT_ADDR	[7:0]						
R1492 (5D4h)	SLIMbus_RX_Ports2				IMRX6_PO	-							IMRX5_PO	-							
D1402	CLIMbus DV Darts?	1		61			[7:0]					CI.			[7:0]						

SLIMRX8_PORT_ADDR [7:0]

SLIMTX2_PORT_ADDR [7:0]

SLIMTX4_PORT_ADDR [7:0]

SLIMTX6_PORT_ADDR [7:0]

SLIMTX8_PORT_ADDR [7:0]



Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R1507 (5E3h)	SLIMbus_Framer_Ref_ Gear	0	0	0	0	0	0	0	0	0	0	0	SLIMCLK_ SRC	. SL	.IMCLK_RE	EF_GEAR [3:0]	0000h
R1509 (5E5h)	SLIMbus_Rates_1		SLIM	IRX2_RATE	[4:0]		0	0	0		SLIN	IRX1_RATE	E [4:0]		0	0	0	0000h
R1510 (5E6h)	SLIMbus_Rates_2		SLIM	IRX4_RATE	[4:0]		0	0	0		SLIN	IRX3_RATE	E [4:0]		0	0	0	0000h
R1511 (5E7h)	SLIMbus_Rates_3		SLIN	IRX6_RATE	[4:0]		0	0	0		SLIN	IRX5_RATE	E [4:0]		0	0	0	0000h
R1512 (5E8h)	SLIMbus_Rates_4		SLIM	IRX8_RATE	[4:0]		0	0	0		SLIN	IRX7_RATE	E [4:0]		0	0	0	0000h
R1513 (5E9h)	SLIMbus_Rates_5		SLIN	ITX2_RATE	[4:0]		0	0	0		SLIN	ITX1_RATE	E [4:0]		0	0	0	0000h
R1514 (5EAh)	SLIMbus_Rates_6		SLIN	ITX4_RATE	[4:0]		0	0	0		SLIN	ITX3_RATE	E [4:0]		0	0	0	0000h
R1515 (5EBh)	SLIMbus_Rates_7		SLIN	ITX6_RATE	[4:0]		0	0	0		SLIN	ITX5_RATE	E [4:0]		0	0	0	0000h
R1516 (5ECh)	SLIMbus_Rates_8		SLIN	ITX8_RATE	[4:0]		0	0	0		SLIN	ITX7_RATE	[4:0]		0	0	0	0000h
R1520 (5F0h)	Slimbus_Pad_Ctrl	0	0	0	0	0	0	0	0	0	0	0	0	0	SLIMDAT2 _DRV_ _STR	SLIMDAT1 _DRV_ _STR	SLIMCLK_ DRV_STR	0007h
R1525 (5F5h)	SLIMbus_RX_Channel_ Enable	0	0	0	0	0	0	0	0	SLIMRX8_ ENA	SLIMRX7_ ENA	SLIMRX6_ ENA	SLIMRX5_ ENA	SLIMRX4_ ENA	SLIMRX3_ ENA	SLIMRX2_ ENA	SLIMRX1_ ENA	0000h
R1526 (5F6h)	SLIMbus_TX_Channel_ Enable	0	0	0	0	0	0	0	0	SLIMTX8_ ENA	SLIMTX7_ ENA	SLIMTX6_ ENA	SLIMTX5_ ENA	SLIMTX4_ ENA	SLIMTX3_ ENA	SLIMTX2_ ENA	SLIMTX1_ ENA	0000h
R1527 (5F7h)	SLIMbus_RX_Port_ Status	0	0	0	0	0	0	0	0	SLIMRX8_ PORT_ STS	SLIMRX7_ PORT_ STS	SLIMRX6_ PORT_ STS	SLIMRX5_ PORT_ STS	SLIMRX4_ PORT_ STS	SLIMRX3_ PORT_ STS	SLIMRX2 PORT_ STS	SLIMRX1_ PORT_ STS	0000h
R1528 (5F8h)	SLIMbus_TX_Port_ Status	0	0	0	0	0	0	0	0	SLIMTX8_ PORT	SLIMTX7_ PORT_	SLIMTX6_ PORT_	SLIMTX5_ PORT_	SLIMTX4_ PORT_	SLIMTX3_ PORT_	SLIMTX2_ PORT_	SLIMTX1_ PORT_	0000h
R1600 (640h)	PWM1MIX_Input_1_ Source	PWM1MIX _STS	0	0	0	0	0	0	0	STS ⁻	STS	STS	STS PWM1MIX	STS _SRC1 [7:0]	STS	STS	STS ⁻	0000h
R1601 (641h)	PWM1MIX_Input_1_ Volume	0	0	0	0	0	0	0	0			PWN	11MIX_VOL	1 [6:0]			0	0080h
R1602 (642h)	PWM1MIX_Input_2_ Source	PWM1MIX _STS	0	0	0	0	0	0	0				PWM1MIX	_SRC2 [7:0]]		I	0000h
R1603 (643h)	PWM1MIX_Input_2_ Volume	0	0	0	0	0	0	0	0			PWN	11MIX_VOL	2 [6:0]			0	0080h
R1604 (644h)	PWM1MIX_Input_3_ Source	PWM1MIX _STS	0	0	0	0	0	0	0				PWM1MIX	_SRC3 [7:0]]			0000h
R1605 (645h)	PWM1MIX_Input_3_ Volume	0	0	0	0	0	0	0	0			PWN	11MIX_VOL	3 [6:0]			0	0080h
R1606 (646h)	PWM1MIX_Input_4_ Source	PWM1MIX _STS	0	0	0	0	0	0	0				PWM1MIX	_SRC4 [7:0]]			0000h
R1607 (647h)	PWM1MIX_Input_4_ Volume	0	0	0	0	0	0	0	0			PWN	11MIX_VOL	4 [6:0]			0	0080h
R1608 (648h)	PWM2MIX_Input_1_ Source	PWM2MIX _STS	0	0	0	0	0	0	0				PWM2MIX	_SRC1 [7:0]		•	0000h
R1609 (649h)	PWM2MIX_Input_1_ Volume	0	0	0	0	0	0	0	0			PWN	12MIX_VOL	1 [6:0]			0	0080h
R1610 (64Ah)	PWM2MIX_Input_2_ Source	PWM2MIX _STS	0	0	0	0	0	0	0				PWM2MIX	_SRC2 [7:0]]			0000h
	PWM2MIX_Input_2_ Volume	0	0	0	0	0	0	0	0			PWN	12MIX_VOL	2 [6:0]			0	0080h
R1612	PWM2MIX_Input_3_ Source	PWM2MIX _STS	0	0	0	0	0	0	0				PWM2MIX	_SRC3 [7:0]			0000h
R1613 (64Dh)	PWM2MIX_Input_3_ Volume	0	0	0	0	0	0	0	0			PWN	12MIX_VOL	3 [6:0]			0	0080h
R1614 (64Eh)	PWM2MIX_Input_4_ Source	PWM2MIX _STS	0	0	0	0	0	0	0				PWM2MIX	_SRC4 [7:0]]			0000h
	PWM2MIX_Input_4_ Volume	0	0	0	0	0	0	0	0			PWN	12MIX_VOL	4 [6:0]			0	0080h
R1664	OUT1LMIX_Input_1_ Source	OUT1LMIX _STS	0	0	0	0	0	0	0				OUT1LMIX	_SRC1 [7:0]			0000h
R1665 (681h)	OUT1LMIX_Input_1_ Volume	0	0	0	0	0	0	0	0			OUT	ILMIX_VOL	.1 [6:0]			0	0080h
R1666 (682h)	OUT1LMIX_Input_2_ Source	OUT1LMIX _STS	0	0	0	0	0	0	0				OUT1LMIX	_SRC2 [7:0]			0000h
R1667 (683h)	OUT1LMIX_Input_2_ Volume	0	0	0	0	0	0	0	0			OUT	ILMIX_VOL	2 [6:0]			0	0080h
R1668	OUT1LMIX_Input_3_ Source	OUT1LMIX _STS	0	0	0	0	0	0	0				OUT1LMIX	_SRC3 [7:0]			0000h
R1669 (685h)	OUT1LMIX_Input_3_ Volume	0	0	0	0	0	0	0	0				ILMIX_VOL				0	0080h
()	OUT1LMIX_Input_4_ Source	OUT1LMIX _STS	0	0	0	0	0	0	0					_SRC4 [7:0]			0000h
R1671 (687h)	OUT1LMIX_Input_4_ Volume	0	0	0	0	0	0	0	0			OUT	ILMIX_VOL	4 [6:0]			0	0080h



Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
R1672 (688h)	OUT1RMIX_Input_1_ Source	OUT1RMI X_STS	0	0	0	0	0	0	0	OUT1RMIX_SRC1 [7:0]	0000h
R1673 (689h)	OUT1RMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT1RMIX_VOL1 [6:0] 0	0080h
R1674 (68Ah)	OUT1RMIX_Input_2_ Source	OUT1RMI X_STS	0	0	0	0	0	0	0	OUT1RMIX_SRC2 [7:0]	0000h
R1675 (68Bh)	OUT1RMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT1RMIX_VOL2 [6:0] 0	0080h
R1676 (68Ch)	OUT1RMIX_Input_3_ Source	OUT1RMI X_STS	0	0	0	0	0	0	0	OUT1RMIX_SRC3 [7:0]	0000h
R1677 (68Dh)	OUT1RMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT1RMIX_VOL3 [6:0] 0	0080h
R1678 (68Eh)	OUT1RMIX_Input_4_ Source	OUT1RMI X_STS	0	0	0	0	0	0	0	OUT1RMIX_SRC4 [7:0]	0000h
R1679 (68Fh)	OUT1RMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT1RMIX_VOL4 [6:0] 0	0080h
R1680 (690h)	OUT2LMIX_Input_1_ Source	OUT2LMIX _STS	0	0	0	0	0	0	0	OUT2LMIX_SRC1 [7:0]	0000h
R1681 (691h)	OUT2LMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT2LMIX_VOL1 [6:0] 0	0080h
R1682 (692h)	OUT2LMIX_Input_2_ Source	OUT2LMIX _STS	0	0	0	0	0	0	0	OUT2LMIX_SRC2 [7:0]	0000h
R1683 (693h)	OUT2LMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT2LMIX_VOL2 [6:0] 0	0080h
R1684 (694h)	OUT2LMIX_Input_3_ Source	OUT2LMIX _STS	0	0	0	0	0	0	0	OUT2LMIX_SRC3 [7:0]	0000h
R1685 (695h)	OUT2LMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT2LMIX_VOL3 [6:0] 0	0080h
R1686 (696h)	OUT2LMIX_Input_4_ Source	OUT2LMIX _STS	0	0	0	0	0	0	0	OUT2LMIX_SRC4 [7:0]	0000h
R1687 (697h)	OUT2LMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT2LMIX_VOL4 [6:0] 0	0080h
R1688 (698h)	OUT2RMIX_Input_1_ Source	OUT2RMI X_STS	0	0	0	0	0	0	0	OUT2RMIX_SRC1 [7:0]	0000h
R1689 (699h)	OUT2RMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT2RMIX_VOL1 [6:0] 0	0080h
R1690 (69Ah)	OUT2RMIX_Input_2_ Source	OUT2RMI X_STS	0	0	0	0	0	0	0	OUT2RMIX_SRC2 [7:0]	0000h
R1691 (69Bh)	OUT2RMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT2RMIX_VOL2 [6:0] 0	0080h
R1692 (69Ch)	OUT2RMIX_Input_3_ Source	OUT2RMI X_STS	0	0	0	0	0	0	0	OUT2RMIX_SRC3 [7:0]	0000h
R1693 (69Dh)	OUT2RMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT2RMIX_VOL3 [6:0] 0	0080h
R1694 (69Eh)	OUT2RMIX_Input_4_ Source	OUT2RMI X_STS	0	0	0	0	0	0	0	OUT2RMIX_SRC4 [7:0]	0000h
R1695 (69Fh)	OUT2RMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT2RMIX_VOL4 [6:0] 0	0080h
R1696 (6A0h)	OUT3LMIX_Input_1_ Source	OUT3LMIX _STS	0	0	0	0	0	0	0	OUT3LMIX_SRC1 [7:0]	0000h
R1697 (6A1h)	OUT3LMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT3LMIX_VOL1 [6:0] 0	0080h
R1698 (6A2h)	OUT3LMIX_Input_2_ Source	OUT3LMIX _STS	0	0	0	0	0	0	0	OUT3LMIX_SRC2 [7:0]	0000h
R1699 (6A3h)	OUT3LMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT3LMIX_VOL2 [6:0] 0	0080h
R1700 (6A4h)	OUT3LMIX_Input_3_ Source	OUT3LMIX _STS	0	0	0	0	0	0	0	OUT3LMIX_SRC3 [7:0]	0000h
R1701 (6A5h)	OUT3LMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT3LMIX_VOL3 [6:0] 0	0080h
R1702 (6A6h)	OUT3LMIX_Input_4_ Source	OUT3LMIX _STS	0	0	0	0	0	0	0	OUT3LMIX_SRC4 [7:0]	0000h
R1703 (6A7h)	OUT3LMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT3LMIX_VOL4 [6:0] 0	0080h
R1704 (6A8h)	OUT3RMIX_Input_1_ Source	OUT3RMI X_STS	0	0	0	0	0	0	0	OUT3RMIX_SRC1 [7:0]	0000h
R1705 (6A9h)	OUT3RMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT3RMIX_VOL1 [6:0] 0	0080h
R1706 (6AAh)	OUT3RMIX_Input_2_ Source	OUT3RMI X_STS	0	0	0	0	0	0	0	OUT3RMIX_SRC2 [7:0]	0000h
R1707 (6ABh)	OUT3RMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT3RMIX_VOL2 [6:0] 0	0080h
R1708 (6ACh)	OUT3RMIX_Input_3_ Source	OUT3RMI X_STS	0	0	0	0	0	0	0	OUT3RMIX_SRC3 [7:0]	0000h
R1709 (6ADh)	OUT3RMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT3RMIX_VOL3 [6:0] 0	0080h



(6AEh) 8					12	11	10	9	8	7 6 5 4 3 2 1 0	Default
	OUT3RMIX_Input_4_ Source	OUT3RMI X STS	0	0	0	0	0	0	0	OUT3RMIX_SRC4 [7:0]	0000h
R1711 (6AFh)	OUT3RMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT3RMIX_VOL4 [6:0] 0	0080h
	OUT5LMIX_Input_1_ Source	OUT5LMIX _STS	0	0	0	0	0	0	0	OUT5LMIX_SRC1 [7:0]	0000h
R1729	OUT5LMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT5LMIX_VOL1 [6:0] 0	0080h
	OUT5LMIX_Input_2_ Source	OUT5LMIX _STS	0	0	0	0	0	0	0	OUT5LMIX_SRC2 [7:0]	0000h
R1731	OUT5LMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT5LMIX_VOL2 [6:0] 0	0080h
R1732	OUT5LMIX_Input_3_ Source	OUT5LMIX _STS	0	0	0	0	0	0	0	OUT5LMIX_SRC3 [7:0]	0000h
	OUT5LMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT5LMIX_VOL3 [6:0] 0	0080h
	OUT5LMIX_Input_4_ Source	OUT5LMIX _STS	0	0	0	0	0	0	0	OUT5LMIX_SRC4 [7:0]	0000h
	OUT5LMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT5LMIX_VOL4 [6:0] 0	0080h
	OUT5RMIX_Input_1_ Source	OUT5RMI X_STS	0	0	0	0	0	0	0	OUT5RMIX_SRC1 [7:0]	0000h
	OUT5RMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	OUT5RMIX_VOL1 [6:0] 0	0080h
(6CAh) 8	OUT5RMIX_Input_2_ Source	OUT5RMI X_STS	0	0	0	0	0	0	0	OUT5RMIX_SRC2 [7:0]	0000h
	OUT5RMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	OUT5RMIX_VOL2 [6:0] 0	0080h
	OUT5RMIX_Input_3_ Source	OUT5RMI X_STS	0	0	0	0	0	0	0	OUT5RMIX_SRC3 [7:0]	0000h
	OUT5RMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	OUT5RMIX_VOL3 [6:0] 0	0080h
(6CEh) 8	OUT5RMIX_Input_4_ Source	OUT5RMI X_STS	0	0	0	0	0	0	0	OUT5RMIX_SRC4 [7:0]	0000h
	OUT5RMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	OUT5RMIX_VOL4 [6:0] 0	0080h
	AIF1TX1MIX_Input_1_ Source	AIF1TX1MI X_STS	0	0	0	0	0	0	0	AIF1TX1MIX_SRC1 [7:0]	0000h
	AIF1TX1MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX1MIX_VOL1 [6:0] 0	0080h
	AIF1TX1MIX_Input_2_ Source	AIF1TX1MI X_STS	0	0	0	0	0	0	0	AIF1TX1MIX_SRC2 [7:0]	0000h
(703h)	AIF1TX1MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX1MIX_VOL2 [6:0] 0	0080h
(704h)	AIF1TX1MIX_Input_3_ Source	AIF1TX1MI X_STS	0	0	0	0	0	0	0	AIF1TX1MIX_SRC3 [7:0]	0000h
(705h)	AIF1TX1MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX1MIX_VOL3 [6:0] 0	0080h
	AIF1TX1MIX_Input_4_ Source	AIF1TX1MI X_STS	0	0	0	0	0	0	0	AIF1TX1MIX_SRC4 [7:0]	0000h
	AIF1TX1MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX1MIX_VOL4 [6:0] 0	0080h
	AIF1TX2MIX_Input_1_ Source	AIF1TX2MI X_STS	0	0	0	0	0	0	0	AIF1TX2MIX_SRC1 [7:0]	0000h
	AIF1TX2MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX2MIX_VOL1 [6:0] 0	0080h
	AIF1TX2MIX_Input_2_ Source	AIF1TX2MI X_STS	0	0	0	0	0	0	0	AIF1TX2MIX_SRC2 [7:0]	0000h
(70Bh)	AIF1TX2MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX2MIX_VOL2 [6:0] 0	0080h
	AIF1TX2MIX_Input_3_ Source	AIF1TX2MI X_STS	0	0	0	0	0	0	0	AIF1TX2MIX_SRC3 [7:0]	0000h
(70Dh)	AIF1TX2MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX2MIX_VOL3 [6:0] 0	0080h
(70Eh)	AIF1TX2MIX_Input_4_ Source	AIF1TX2MI X_STS	0	0	0	0	0	0	0	AIF1TX2MIX_SRC4 [7:0]	0000h
(70Fh)	AIF1TX2MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX2MIX_VOL4 [6:0] 0	0080h
(710h)	AIF1TX3MIX_Input_1_ Source	AIF1TX3MI X_STS	0	0	0	0	0	0	0	AIF1TX3MIX_SRC1 [7:0]	0000h
(711h)	AIF1TX3MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX3MIX_VOL1 [6:0] 0	0080h
(712h)	AIF1TX3MIX_Input_2_ Source	AIF1TX3MI X_STS	0	0	0	0	0	0	0	AIF1TX3MIX_SRC2 [7:0]	0000h
	AIF1TX3MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX3MIX_VOL2 [6:0] 0	0080h



Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
R1812 (714h)	AIF1TX3MIX_Input_3_ Source	AIF1TX3MI X_STS	0	0	0	0	0	0	0	AIF1TX3MIX_SRC3 [7:0]	0000h
R1813 (715h)	AIF1TX3MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX3MIX_VOL3 [6:0] 0	0080h
R1814 (716h)	AIF1TX3MIX_Input_4_ Source	AIF1TX3MI X_STS	0	0	0	0	0	0	0	AIF1TX3MIX_SRC4 [7:0]	0000h
R1815 (717h)	AIF1TX3MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX3MIX_VOL4 [6:0] 0	0080h
R1816 (718h)	AIF1TX4MIX_Input_1_ Source	AIF1TX4MI X_STS	0	0	0	0	0	0	0	AIF1TX4MIX_SRC1 [7:0]	0000h
R1817 (719h)	AIF1TX4MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX4MIX_VOL1 [6:0] 0	0080h
R1818 (71Ah)	AIF1TX4MIX_Input_2_ Source	AIF1TX4MI X_STS	0	0	0	0	0	0	0	AIF1TX4MIX_SRC2 [7:0]	0000h
R1819 (71Bh)	AIF1TX4MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX4MIX_VOL2 [6:0] 0	0080h
R1820 (71Ch)	AIF1TX4MIX_Input_3_ Source	AIF1TX4MI X_STS	0	0	0	0	0	0	0	AIF1TX4MIX_SRC3 [7:0]	0000h
R1821 (71Dh)	AIF1TX4MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX4MIX_VOL3 [6:0] 0	0080h
R1822 (71Eh)	AIF1TX4MIX_Input_4_ Source	AIF1TX4MI X_STS	0	0	0	0	0	0	0	AIF1TX4MIX_SRC4 [7:0]	0000h
R1823 (71Fh)	AIF1TX4MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX4MIX_VOL4 [6:0] 0	0080h
R1824 (720h)	AIF1TX5MIX_Input_1_ Source	AIF1TX5MI X_STS	0	0	0	0	0	0	0	AIF1TX5MIX_SRC1 [7:0]	0000h
R1825 (721h)	AIF1TX5MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX5MIX_VOL1 [6:0] 0	0080h
R1826 (722h)	AIF1TX5MIX_Input_2_ Source	AIF1TX5MI X_STS	0	0	0	0	0	0	0	AIF1TX5MIX_SRC2 [7:0]	0000h
R1827 (723h)	AIF1TX5MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX5MIX_VOL2 [6:0] 0	0080h
R1828 (724h)	AIF1TX5MIX_Input_3_ Source	AIF1TX5MI X_STS	0	0	0	0	0	0	0	AIF1TX5MIX_SRC3 [7:0]	0000h
R1829 (725h)	AIF1TX5MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX5MIX_VOL3 [6:0] 0	0080h
R1830 (726h)	AIF1TX5MIX_Input_4_ Source	AIF1TX5MI X_STS	0	0	0	0	0	0	0	AIF1TX5MIX_SRC4 [7:0]	0000h
R1831 (727h)	AIF1TX5MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX5MIX_VOL4 [6:0] 0	0080h
R1832 (728h)	AIF1TX6MIX_Input_1_ Source	AIF1TX6MI X_STS	0	0	0	0	0	0	0	AIF1TX6MIX_SRC1 [7:0]	0000h
R1833 (729h)	AIF1TX6MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX6MIX_VOL1 [6:0] 0	0080h
R1834 (72Ah)	AIF1TX6MIX_Input_2_ Source	AIF1TX6MI X_STS	0	0	0	0	0	0	0	AIF1TX6MIX_SRC2 [7:0]	0000h
R1835 (72Bh)	AIF1TX6MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX6MIX_VOL2 [6:0] 0	0080h
R1836 (72Ch)	AIF1TX6MIX_Input_3_ Source	AIF1TX6MI X_STS	0	0	0	0	0	0	0	AIF1TX6MIX_SRC3 [7:0]	0000h
R1837 (72Dh)	AIF1TX6MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX6MIX_VOL3 [6:0] 0	0080h
R1838 (72Eh)	AIF1TX6MIX_Input_4_ Source	AIF1TX6MI X_STS	0	0	0	0	0	0	0	AIF1TX6MIX_SRC4 [7:0]	0000h
R1839 (72Fh)	AIF1TX6MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX6MIX_VOL4 [6:0] 0	0080h
R1840 (730h)	AIF1TX7MIX_Input_1_ Source	AIF1TX7MI X_STS	0	0	0	0	0	0	0	AIF1TX7MIX_SRC1 [7:0]	0000h
R1841 (731h)	AIF1TX7MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX7MIX_VOL1 [6:0] 0	0080h
R1842 (732h)	AIF1TX7MIX_Input_2_ Source	AIF1TX7MI X_STS	0	0	0	0	0	0	0	AIF1TX7MIX_SRC2 [7:0]	0000h
R1843 (733h)	AIF1TX7MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX7MIX_VOL2 [6:0] 0	0080h
R1844 (734h)	AIF1TX7MIX_Input_3_ Source	AIF1TX7MI X_STS	0	0	0	0	0	0	0	AIF1TX7MIX_SRC3 [7:0]	0000h
R1845 (735h)	AIF1TX7MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX7MIX_VOL3 [6:0] 0	0080h
R1846 (736h)	AIF1TX7MIX_Input_4_ Source	AIF1TX7MI X_STS	0	0	0	0	0	0	0	AIF1TX7MIX_SRC4 [7:0]	0000h
R1847 (737h)	AIF1TX7MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX7MIX_VOL4 [6:0] 0	0080h
R1848 (738h)	AIF1TX8MIX_Input_1_ Source	AIF1TX8MI X_STS	0	0	0	0	0	0	0	AIF1TX8MIX_SRC1 [7:0]	0000h
R1849 (739h)	AIF1TX8MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF1TX8MIX_VOL1 [6:0] 0	0080h



Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
R1850 (73Ah)	AIF1TX8MIX_Input_2_ Source	AIF1TX8MI X_STS	0	0	0	0	0	0	0	AIF1TX8MIX_SRC2 [7:0]	0000h
R1851 (73Bh)	AIF1TX8MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF1TX8MIX_VOL2 [6:0] 0	0080h
R1852 (73Ch)	AIF1TX8MIX_Input_3_ Source	AIF1TX8MI X_STS	0	0	0	0	0	0	0	AIF1TX8MIX_SRC3 [7:0]	0000h
R1853 (73Dh)	AIF1TX8MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF1TX8MIX_VOL3 [6:0] 0	0080h
R1854 (73Eh)	AIF1TX8MIX_Input_4_ Source	AIF1TX8MI X_STS	0	0	0	0	0	0	0	AIF1TX8MIX_SRC4 [7:0]	0000h
R1855 (73Fh)	AIF1TX8MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF1TX8MIX_VOL4 [6:0] 0	0080h
R1856 (740h)	AIF2TX1MIX_Input_1_ Source	AIF2TX1MI X_STS	0	0	0	0	0	0	0	AIF2TX1MIX_SRC1 [7:0]	0000h
R1857 (741h)	AIF2TX1MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF2TX1MIX_VOL1 [6:0] 0	0080h
R1858 (742h)	AIF2TX1MIX_Input_2_ Source	AIF2TX1MI X_STS	0	0	0	0	0	0	0	AIF2TX1MIX_SRC2 [7:0]	0000h
R1859 (743h)	AIF2TX1MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF2TX1MIX_VOL2 [6:0] 0	0080h
R1860 (744h)	AIF2TX1MIX_Input_3_ Source	AIF2TX1MI X_STS	0	0	0	0	0	0	0	AIF2TX1MIX_SRC3 [7:0]	0000h
R1861 (745h)	AIF2TX1MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF2TX1MIX_VOL3 [6:0] 0	0080h
R1862 (746h)	AIF2TX1MIX_Input_4_ Source	AIF2TX1MI X_STS	0	0	0	0	0	0	0	AIF2TX1MIX_SRC4 [7:0]	0000h
R1863 (747h)	AIF2TX1MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF2TX1MIX_VOL4 [6:0] 0	0080h
R1864 (748h)	AIF2TX2MIX_Input_1_ Source	AIF2TX2MI X_STS	0	0	0	0	0	0	0	AIF2TX2MIX_SRC1 [7:0]	0000h
R1865 (749h)	AIF2TX2MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF2TX2MIX_VOL1 [6:0] 0	0080h
R1866 (74Ah)	AIF2TX2MIX_Input_2_ Source	AIF2TX2MI X_STS	0	0	0	0	0	0	0	AIF2TX2MIX_SRC2 [7:0]	0000h
R1867 (74Bh)	AIF2TX2MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF2TX2MIX_VOL2 [6:0] 0	0080h
R1868 (74Ch)	AIF2TX2MIX_Input_3_ Source	AIF2TX2MI X_STS	0	0	0	0	0	0	0	AIF2TX2MIX_SRC3 [7:0]	0000h
R1869 (74Dh)	AIF2TX2MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF2TX2MIX_VOL3 [6:0] 0	0080h
R1870 (74Eh)	AIF2TX2MIX_Input_4_ Source	AIF2TX2MI X_STS	0	0	0	0	0	0	0	AIF2TX2MIX_SRC4 [7:0]	0000h
R1871 (74Fh)	AIF2TX2MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF2TX2MIX_VOL4 [6:0] 0	0080h
R1872 (750h)	AIF2TX3MIX_Input_1_ Source	AIF2TX3MI X_STS	0	0	0	0	0	0	0	AIF2TX3MIX_SRC1 [7:0]	0000h
R1873 (751h)	AIF2TX3MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF2TX3MIX_VOL1 [6:0] 0	0080h
R1874 (752h)	AIF2TX3MIX_Input_2_ Source	AIF2TX3MI X_STS	0	0	0	0	0	0	0	AIF2TX3MIX_SRC2 [7:0]	0000h
R1875 (753h)	AIF2TX3MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF2TX3MIX_VOL2 [6:0] 0	0080h
R1876 (754h)	AIF2TX3MIX_Input_3_ Source	AIF2TX3MI X_STS	0	0	0	0	0	0	0	AIF2TX3MIX_SRC3 [7:0]	0000h
R1877 (755h)	AIF2TX3MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF2TX3MIX_VOL3 [6:0] 0	0080h
R1878 (756h)	AIF2TX3MIX_Input_4_ Source	AIF2TX3MI X_STS	0	0	0	0	0	0	0	AIF2TX3MIX_SRC4 [7:0]	0000h
R1879 (757h)	AIF2TX3MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF2TX3MIX_VOL4 [6:0] 0	0080h
R1880 (758h)	AIF2TX4MIX_Input_1_ Source	AIF2TX4MI X_STS	0	0	0	0	0	0	0	AIF2TX4MIX_SRC1 [7:0]	0000h
R1881 (759h)	AIF2TX4MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF2TX4MIX_VOL1 [6:0] 0	0080h
R1882 (75Ah)	AIF2TX4MIX_Input_2_ Source	AIF2TX4MI X_STS	0	0	0	0	0	0	0	AIF2TX4MIX_SRC2 [7:0]	0000h
R1883 (75Bh)	AIF2TX4MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF2TX4MIX_VOL2 [6:0] 0	0080h
R1884 (75Ch)	AIF2TX4MIX_Input_3_ Source	AIF2TX4MI X_STS	0	0	0	0	0	0	0	AIF2TX4MIX_SRC3 [7:0]	0000h
R1885 (75Dh)	AIF2TX4MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF2TX4MIX_VOL3 [6:0] 0	0080h
R1886 (75Eh)	AIF2TX4MIX_Input_4_ Source	AIF2TX4MI X_STS	0	0	0	0	0	0	0	AIF2TX4MIX_SRC4 [7:0]	0000h
(75Eh)	AIF2TX4MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF2TX4MIX_VOL4 [6:0] 0	0080h
(1011)	Volumo	1		1			I	1	1		



Heads Marger Statur, Imp.L. Marger Stat	Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
HERE PATTORNAL (Ppd. 1) J J Q Q Q Q Q Q D PATTORNAL (Ppd. 2) J O OSE HERE PATTORNAL (Ppd. 2) V Q				0	0	0	0	0	0	0	AIF2TX5MIX_SRC1 [7:0]	0000h
(TRD) Boards 7,18 1	R1889	AIF2TX5MIX_Input_1_	-	0	0	0	0	0	0	0	AIF2TX5MIX_VOL1 [6:0] 0	0080h
Heart Matrix Mark Lingul, Z. II D D II D			AIF2TX5MI X_STS	0	0	0	0	0	0	0	AIF2TX5MIX_SRC2 [7:0]	0000h
TOPAH Bourse X.378 Image: Mark Sector 1 Control 1 <thcontrol 1<="" th=""> <thcontrol 1<="" th=""> <thc< td=""><td>R1891</td><td>AIF2TX5MIX_Input_2_</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>AIF2TX5MIX_VOL2 [6:0] 0</td><td>0080h</td></thc<></thcontrol></thcontrol>	R1891	AIF2TX5MIX_Input_2_	0	0	0	0	0	0	0	0	AIF2TX5MIX_VOL2 [6:0] 0	0080h
THISD MP27XMMX_hpud_3 0			AIF2TX5MI X_STS	0	0	0	0	0	0	0	AIF2TX5MIX_SRC3 [7:0]	0000h
High High <th< td=""><td>R1893</td><td>AIF2TX5MIX_Input_3_</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>AIF2TX5MIX_VOL3 [6:0] 0</td><td>0080h</td></th<>	R1893	AIF2TX5MIX_Input_3_	0	0	0	0	0	0	0	0	AIF2TX5MIX_VOL3 [6:0] 0	0080h
HitsS MIZZXMMX Input. 4 0 0 0 0 0 0 AP77MARX_VOLEGR 0	R1894	AIF2TX5MIX_Input_4_	AIF2TX5MI X_STS	0	0	0	0	0	0	0	AIF2TX5MIX_SRC4 [7:0]	0000h
TOBB: Source V.373 Image: Constraint of the source of	R1895		0	0	0	0	0	0	0	0	AIF2TX5MIX_VOL4 [6:0] 0	0080h
(76%) Notime Martine	R1896			0	0	0	0	0	0	0	AIF2TX6MIX_SRC1 [7:0]	0000h
Creating Source X.STS Image: Control of the second sec	R1897		0	0	0	0	0	0	0	0	AIF2TX6MIX_VOL1 [6:0] 0	0080h
(768b) Notime - <th< td=""><td></td><td></td><td>AIF2TX6MI X_STS</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>AIF2TX6MIX_SRC2 [7:0]</td><td>0000h</td></th<>			AIF2TX6MI X_STS	0	0	0	0	0	0	0	AIF2TX6MIX_SRC2 [7:0]	0000h
RTROD MEZTXARX Control C C C C C APETXARX CALL COND COND <thcnt< td=""><td>R1899</td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>AIF2TX6MIX_VOL2 [6:0] 0</td><td>0080h</td></thcnt<>	R1899		0	0	0	0	0	0	0	0	AIF2TX6MIX_VOL2 [6:0] 0	0080h
RT601 AIP2TXMMX_PUPU_3 0 0 0 0 0 0 AIP2TXMMX_VOL3[60] 0 0000h RT602 AIP2TXMMX_PUPU_4 NETXMM 0<	R1900		AIF2TX6MI X_STS	0	0	0	0	0	0	0	AIF2TX6MIX_SRC3 [7:0]	0000h
(PTED) Source X.57	R1901		0	0	0	0	0	0	0	0	AIF2TX6MIX_VOL3 [6:0] 0	0080h
(7PFh) Volume (1) (R1902			0	0	0	0	0	0	0	AIF2TX6MIX_SRC4 [7:0]	0000h
R1904 Alf217X/ML (nput_1XSTS 0 0 0 0 0 0 0 Alf217X/ML (nput_1) 0 0 0 0 0 Alf217X/ML (nput_1) 0 0 0 0 0 0 0 0 0 Alf217X/ML (nput_1) 0 0 0 0 0 0 0 Alf217X/ML (nput_1) 0 0 0 0 0 0 0 Alf217X/ML (nput_2) 0 0 0 0 0 0 0 0 0 Alf217X/ML (nput_1) 0 <td>R1903</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>AIF2TX6MIX_VOL4 [6:0] 0</td> <td>0080h</td>	R1903		0	0	0	0	0	0	0	0	AIF2TX6MIX_VOL4 [6:0] 0	0080h
R1906 AF2TX7MX_Input_1 0 0 0 0 0 AF2TX7MX_V[0] 0 0000h R1906 AF2TX7MX_Input_2 W2TX7MM 0 0 0 0 0 AF2TX7MX_V[0] 0 0 0 0 AF2TX7MX_V[0] 0 0 0 0 0 AF2TX7MX_V[0] 0 0 0 0 0 0 0 AF2TX7MX_V[0] 0	R1904		AIF2TX7MI X_STS	0	0	0	0	0	0	0	AIF2TX7MIX_SRC1 [7:0]	0000h
R1906 AFE2TX/MIX_Input_2_ View 0 0 0 0 0 0 AFE2TX/MIX_SRC2[7:0] 0 0000h R1907 AFE2TX/MIX_Input_3_ 0	R1905		0	0	0	0	0	0	0	0	AIF2TX7MIX_VOL1 [6:0] 0	0080h
(773) Volume Volume Volume Volume Volume 0000h R1908 AlF2TX7MX_input_3_ VEXTM 0 0 0 0 0 AlF2TX7MX_SRC3[7:0] 0000h R1908 AlF2TX7MX_input_3_ VEXTM 0<			AIF2TX7MI X_STS	0	0	0	0	0	0	0	AIF2TX7MIX_SRC2 [7:0]	0000h
R1908 MF2TX7MIX_Input_3_ MF2TX7MIX_SRC3[7:0] 0000h (774h) Source X_STS 0 0 0 0 0 AlF2TX7MIX_Input_3_ 0			0	0	0	0	0	0	0	0	AIF2TX7MIX_VOL2 [6:0] 0	0080h
(775h) Volume				0	0	0	0	0	0	0	AIF2TX7MIX_SRC3 [7:0]	0000h
R1910 AIF2TX7MIX_Input_4XSTS V 0	R1909		0	0	0	0	0	0	0	0	AIF2TX7MIX_VOL3 [6:0] 0	0080h
(777h) Volume Volume Volume Volume Volume Volume 0	R1910			0	0	0	0	0	0	0	AIF2TX7MIX_SRC4 [7:0]	0000h
(776h) Source X_STS Image: Constraint of the state of the s			0	0	0	0	0	0	0	0	AIF2TX7MIX_VOL4 [6:0] 0	0080h
(779h) Volume Image: Constraint of the second seco			AIF2TX8MI X_STS	0	0	0	0	0	0	0	AIF2TX8MIX_SRC1 [7:0]	0000h
R1914 AIF2TX8MIX_Input_2 AIF2TX8MIX_SRC2 [7:0] 0000h R1914 AIF2TX8MIX_Input_2 0			0	0	0	0	0	0	0	0	AIF2TX8MIX_VOL1 [6:0] 0	0080h
R1915 AIF2TX8MIX_Input_2_ 0	R1914	AIF2TX8MIX_Input_2_	AIF2TX8MI X_STS	0	0	0	0	0	0	0	AIF2TX8MIX_SRC2 [7:0]	0000h
R1916 AIF2TX8MIX_Input_3_ AIF2TX8MIX_Input_3_ AIF2TX8MIX_Input_3_ 0	R1915		0	0	0	0	0	0	0	0	AIF2TX8MIX_VOL2 [6:0] 0	0080h
R1917 (77Dh) AIF2TX8MIX_Input_3_ Volume 0			AIF2TX8MI X_STS	0	0	0	0	0	0	0	AIF2TX8MIX_SRC3 [7:0]	0000h
R1918 AIF2TX8MIX_Input_4_ AIF2TX8MIX_STS 0 0 0 0 0 0 AIF2TX8MIX_SRC4 [7:0] 0000h R1919 AIF2TX8MIX_Input_4_ 0 <td>R1917</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>AIF2TX8MIX_VOL3 [6:0] 0</td> <td>0080h</td>	R1917		0	0	0	0	0	0	0	0	AIF2TX8MIX_VOL3 [6:0] 0	0080h
R1919 AIF2TX8MIX_Input_4 0	R1918	AIF2TX8MIX_Input_4_		0	0	0	0	0	0	0	AIF2TX8MIX_SRC4 [7:0]	0000h
R1920 (780h) AIF3TX1MIX_Input_1_ Source AIF3TX1MIX_Input_1_ X_STS 0	R1919	AIF2TX8MIX_Input_4_	0	0	0	0	0	0	0	0	AIF2TX8MIX_VOL4 [6:0] 0	0080h
R1921 AIF3TX1MIX_Input_1_ 0	R1920	AIF3TX1MIX_Input_1_	AIF3TX1MI X_STS	0	0	0	0	0	0	0	AIF3TX1MIX_SRC1 [7:0]	0000h
R1922 (782h) AIF3TX1MIX_Input_2 Source AIF3TX1MIX_Input_2 X_STS 0	R1921	AIF3TX1MIX_Input_1_	0	0	0	0	0	0	0	0	AIF3TX1MIX_VOL1 [6:0] 0	0080h
R1923 AIF3TX1MIX_Input_2 0	R1922	AIF3TX1MIX_Input_2_	AIF3TX1MI X_STS	0	0	0	0	0	0	0	AIF3TX1MIX_SRC2 [7:0]	0000h
R1924 AIF3TX1MIX_Input_3_ AIF3TX1MIX_Input_3_ AIF3TX1MIX_SRC3 [7:0] 0000h (784h) Source 0 0 0 0 0 0 0 0000h R1925 AIF3TX1MIX_Input_3_ 0 0 0 0 0 0 0 0000h	R1923	AIF3TX1MIX_Input_2_	0	0	0	0	0	0	0	0	AIF3TX1MIX_VOL2 [6:0] 0	0080h
R1925 AIF3TX1MIX_Input_3_ 0 0 0 0 0 0 0 0 0 0 AIF3TX1MIX_VOL3[6:0] 0 0080h	R1924	AIF3TX1MIX_Input_3_		0	0	0	0	0	0	0	AIF3TX1MIX_SRC3 [7:0]	0000h
			0	0	0	0	0	0	0	0	AIF3TX1MIX_VOL3 [6:0] 0	0080h



Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
R1926 (786h)	AIF3TX1MIX_Input_4_ Source	AIF3TX1MI X_STS	0	0	0	0	0	0	0	AIF3TX1MIX_SRC4 [7:0]	0000h
R1927 (787h)	AIF3TX1MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF3TX1MIX_VOL4 [6:0] 0	0080h
R1928 (788h)	AIF3TX2MIX_Input_1_ Source	AIF3TX2MI X_STS	0	0	0	0	0	0	0	AIF3TX2MIX_SRC1 [7:0]	0000h
R1929 (789h)	AIF3TX2MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF3TX2MIX_VOL1 [6:0] 0	0080h
R1930 (78Ah)	AIF3TX2MIX_Input_2_ Source	AIF3TX2MI X_STS	0	0	0	0	0	0	0	AIF3TX2MIX_SRC2 [7:0]	0000h
R1931 (78Bh)	AIF3TX2MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF3TX2MIX_VOL2 [6:0] 0	0080h
R1932 (78Ch)	AIF3TX2MIX_Input_3_ Source	AIF3TX2MI X_STS	0	0	0	0	0	0	0	AIF3TX2MIX_SRC3 [7:0]	0000h
R1933 (78Dh)	AIF3TX2MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF3TX2MIX_VOL3 [6:0] 0	0080h
R1934 (78Eh)	AIF3TX2MIX_Input_4_ Source	AIF3TX2MI X_STS	0	0	0	0	0	0	0	AIF3TX2MIX_SRC4 [7:0]	0000h
R1935 (78Fh)	AIF3TX2MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF3TX2MIX_VOL4 [6:0] 0	0080h
R1936 (790h)	AIF3TX3MIX_Input_1_ Source	AIF3TX3MI X_STS	0	0	0	0	0	0	0	AIF3TX3MIX_SRC1 [7:0]	0000h
R1937 (791h)	AIF3TX3MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF3TX3MIX_VOL1 [6:0] 0	0080h
R1938 (792h)	AIF3TX3MIX_Input_2_ Source	AIF3TX3MI X_STS	0	0	0	0	0	0	0	AIF3TX3MIX_SRC2 [7:0]	0000h
R1939 (793h)	AIF3TX3MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF3TX3MIX_VOL2 [6:0] 0	0080h
R1940 (794h)	AIF3TX3MIX_Input_3_ Source	AIF3TX3MI X_STS	0	0	0	0	0	0	0	AIF3TX3MIX_SRC3 [7:0]	0000h
R1941 (795h)	AIF3TX3MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF3TX3MIX_VOL3 [6:0] 0	0080h
R1942 (796h)	AIF3TX3MIX_Input_4_ Source	AIF3TX3MI X_STS	0	0	0	0	0	0	0	AIF3TX3MIX_SRC4 [7:0]	0000h
R1943 (797h)	AIF3TX3MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF3TX3MIX_VOL4 [6:0] 0	0080h
R1944 (798h)	AIF3TX4MIX_Input_1_ Source	AIF3TX4MI X_STS	0	0	0	0	0	0	0	AIF3TX4MIX_SRC1 [7:0]	0000h
R1945 (799h)	AIF3TX4MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF3TX4MIX_VOL1 [6:0] 0	0080h
R1946 (79Ah)	AIF3TX4MIX_Input_2_ Source	AIF3TX4MI X_STS	0	0	0	0	0	0	0	AIF3TX4MIX_SRC2 [7:0]	0000h
R1947 (79Bh)	AIF3TX4MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF3TX4MIX_VOL2 [6:0] 0	0080h
R1948 (79Ch)	AIF3TX4MIX_Input_3_ Source	AIF3TX4MI X_STS	0	0	0	0	0	0	0	AIF3TX4MIX_SRC3 [7:0]	0000h
R1949 (79Dh)	AIF3TX4MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF3TX4MIX_VOL3 [6:0] 0	0080h
R1950 (79Eh)	AIF3TX4MIX_Input_4_ Source	AIF3TX4MI X_STS	0	0	0	0	0	0	0	AIF3TX4MIX_SRC4 [7:0]	0000h
R1951 (79Fh)	AIF3TX4MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF3TX4MIX_VOL4 [6:0] 0	0080h
R1952 (7A0h)	AIF3TX5MIX_Input_1_ Source	AIF3TX5MI X_STS	0	0	0	0	0	0	0	AIF3TX5MIX_SRC1 [7:0]	0000h
R1953 (7A1h)	AIF3TX5MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF3TX5MIX_VOL1 [6:0] 0	0080h
R1954 (7A2h)	AIF3TX5MIX_Input_2_ Source	AIF3TX5MI X_STS	0	0	0	0	0	0	0	AIF3TX5MIX_SRC2 [7:0]	0000h
R1955 (7A3h)	AIF3TX5MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF3TX5MIX_VOL2 [6:0] 0	0080h
R1956 (7A4h)	AIF3TX5MIX_Input_3_ Source	AIF3TX5MI X_STS	0	0	0	0	0	0	0	AIF3TX5MIX_SRC3 [7:0]	0000h
R1957 (7A5h)	AIF3TX5MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	AIF3TX5MIX_VOL3 [6:0] 0	0080h
R1958 (7A6h)	AIF3TX5MIX_Input_4_ Source	AIF3TX5MI X_STS	0	0	0	0	0	0	0	AIF3TX5MIX_SRC4 [7:0]	0000h
R1959 (7A7h)	AIF3TX5MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	AIF3TX5MIX_VOL4 [6:0] 0	0080h
R1960 (7A8h)	AIF3TX6MIX_Input_1_ Source	AIF3TX6MI X_STS	0	0	0	0	0	0	0	AIF3TX6MIX_SRC1 [7:0]	0000h
R1961 (7A9h)	AIF3TX6MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	AIF3TX6MIX_VOL1 [6:0] 0	0080h
R1962 (7AAh)	AIF3TX6MIX_Input_2_ Source	AIF3TX6MI X_STS	0	0	0	0	0	0	0	AIF3TX6MIX_SRC2 [7:0]	0000h
R1963 (7ABh)	AIF3TX6MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	AIF3TX6MIX_VOL2 [6:0] 0	0080h



TRIGE MESTABLY, Input, S. APPROVAL O 0 <th< th=""><th></th><th>1</th><th></th><th></th><th></th><th></th><th></th><th></th><th>1</th><th></th><th></th><th></th></th<>		1							1			
CACO, Botto C.F.5 I <thi< th=""> I I</thi<>	Register	Name		14	13	12	11	10	9	8		Default
(TAD) Nume Prob I </td <td></td> <td>Source</td> <td>X_STS</td> <td>U</td> <td>U</td> <td>0</td> <td>U</td> <td>U</td> <td>U</td> <td>0</td> <td></td> <td>000011</td>		Source	X_STS	U	U	0	U	U	U	0		000011
(TAR) Bourse A.73 Image: An intermediate of the int	(7ADh)	Volume										0080h
(TAP) Nume Num Num< Num< <th< td=""><td></td><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>AIF3TX6MIX_SRC4 [7:0]</td><td>0000h</td></th<>				0	0	0	0	0	0	0	AIF3TX6MIX_SRC4 [7:0]	0000h
(TBD) Source X.87 0 <			0	0	0	0	0	0	0	0	AIF3TX6MIX_VOL4 [6:0] 0	0080h
Fitted MF3TX MIX_[reput_1 0			AIF3TX7MI X_STS	0	0	0	0	0	0	0	AIF3TX7MIX_SRC1 [7:0]	0000h
THOP METRY MML [npul_2] WETRY ML METRY MML [npul_2] WETRY ML METRY MML [npul_2] WETRY ML METRY MML [npul_2] METRY MML [npul	R1969		0	0	0	0	0	0	0	0	AIF3TX7MIX_VOL1 [6:0] 0	0080h
Hight Market Description 0	R1970	AIF3TX7MIX_Input_2_		0	0	0	0	0	0	0	AIF3TX7MIX_SRC2 [7:0]	0000h
THT2 HF3TX/MK_IppU_3_ HF3TX/MK_IpU_13_ 0 0 0 0 0 AH3TX/MK_SRG170 0000h THT3 MF3TX/MK_IpU_3_ 0 0 0 0 0 AH3TX/MK_SRG170 0000h TR137 MF3TX/MK_IpU_3_ 0	R1971	AIF3TX7MIX_Input_2_	0	0	0	0	0	0	0	0	AIF3TX7MIX_VOL2 [6:0] 0	0080h
THT07 MF3TXMMX_IPUL3_ 0 0 0 0 0 0 AF3TXMMX_VOL3(6) 0	R1972	AIF3TX7MIX_Input_3_	AIF3TX7MI X STS	0	0	0	0	0	0	0	AIF3TX7MIX_SRC3 [7:0]	0000h
Fittor AF3TX2MLX_INDUL_4_ M13TX1M 0 0 0 0 0 AF3TX2MLX_INV_CVL [6] 0000h Fittors AF3TX2MLX_INV_CVL [6] 0 0 0 0 AF3TX2MLX_IVVL [6] 0 0 0 0 0 0 AF3TX2MLX_IVVL [6] 0 0 0 0 0 0 AF3TX2MLX_IVVL [6] 0	R1973	AIF3TX7MIX_Input_3_	0	0	0	0	0	0	0	0	AIF3TX7MIX_VOL3 [6:0] 0	0080h
THOSE NESTSCHILK, Input_1 0 0 0 0 0 APSTCHILK, VOL (60) 0 0 0 APSTCHILK, VOL (60) 0 0 0 0 0 0 0 0 APSTCHILK, VOL (60) 0	R1974	AIF3TX7MIX_Input_4_		0	0	0	0	0	0	0	AIF3TX7MIX_SRC4 [7:0]	0000h
F1767 AIF3TX8MX_PUPU_1_ ME3TX8MX_VDL1701 0 0 0 0 0 0 0 AIF3TX8MX_SRC1701 00000h R1977 AIF3TX8MX_PUPU_1_ 0 0 0 0 0 AIF3TX8MX_SRC1701 0	R1975	AIF3TX7MIX_Input_4_	-	0	0	0	0	0	0	0	AIF3TX7MIX_VOL4 [6:0] 0	0080h
RTOP JUST SMUK (Ppul_1, 1) 0 0 0 0 0 AFST SMUK (Ppul_2, 1) 0 <td>R1976</td> <td>AIF3TX8MIX_Input_1_</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>AIF3TX8MIX_SRC1 [7:0]</td> <td>0000h</td>	R1976	AIF3TX8MIX_Input_1_		0	0	0	0	0	0	0	AIF3TX8MIX_SRC1 [7:0]	0000h
R1976 ALF3TX8MX_HUNDL_2	R1977	AIF3TX8MIX_Input_1_	-	0	0	0	0	0	0	0	AIF3TX8MIX_VOL1 [6:0] 0	0080h
Program LFGTXGMX_Input_2_ 0 0 0 0 0 AFSTXBMX_VOL2[6:0] 0 0 00080h R1980 ALFSTXBMX_Input_3_ NFSTXBMX_VOL2[6:0] 0 </td <td>R1978</td> <td>AIF3TX8MIX_Input_2_</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>AIF3TX8MIX_SRC2 [7:0]</td> <td>0000h</td>	R1978	AIF3TX8MIX_Input_2_		0	0	0	0	0	0	0	AIF3TX8MIX_SRC2 [7:0]	0000h
R1990 AF3TX6MX_INput_3X5T3 V3_5T3	R1979	AIF3TX8MIX_Input_2_	-	0	0	0	0	0	0	0	AIF3TX8MIX_VOL2 [6:0] 0	0080h
N = 100000000000000000000000000000000000	R1980	AIF3TX8MIX_Input_3_	AIF3TX8MI	0	0	0	0	0	0	0	AIF3TX8MIX_SRC3 [7:0]	0000h
R1962 JLF3TX8MIX_Input_4_ Mixt SRC [7:0] 0000h (78Eh) Source 0 0 0 0 AF3TX8MIX_SRC4 [7:0] 0000h (78Eh) Source 0 </td <td>R1981</td> <td>AIF3TX8MIX_Input_3_</td> <td>_</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>AIF3TX8MIX_VOL3 [6:0] 0</td> <td>0080h</td>	R1981	AIF3TX8MIX_Input_3_	_	0	0	0	0	0	0	0	AIF3TX8MIX_VOL3 [6:0] 0	0080h
Line D <thd< th=""> D D D</thd<>	R1982	AIF3TX8MIX_Input_4_	AIF3TX8MI	0	0	0	0	0	0	0	AIF3TX8MIX_SRC4 [7:0]	0000h
Preseit SLIMTX-INIX_INPUT_1 SLIMTX-INIX_SRC1 [7:0] 0000h (TCDh) Source IX_STS 0 0 0 0 SLIMTX-INIX_SRC1 [7:0] 0000h (TCDh) Source IX_STS 0	R1983	AIF3TX8MIX_Input_4_	-	0	0	0	0	0	0	0	AIF3TX8MIX_VOL4 [6:0] 0	0080h
Construct Construct <thconstruct< th=""> <thconstruct< th=""> <thc< td=""><td>R1984</td><td>SLIMTX1MIX_Input_1_</td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>SLIMTX1MIX_SRC1 [7:0]</td><td>0000h</td></thc<></thconstruct<></thconstruct<>	R1984	SLIMTX1MIX_Input_1_		0	0	0	0	0	0	0	SLIMTX1MIX_SRC1 [7:0]	0000h
R1986 SLIMTX1MIX_Input_2_ W_STS SLIMTX1MIX_Input_2_ W_STS SLIMTX1MIX_Input_2_ W_STS SLIMTX1MIX_Input_2_ W_STS SLIMTX1MIX_Input_2_ W_STS SLIMTX1MIX_SRC3[7:0] 0000h R1987 SLIMTX1MIX_Input_3_ W_STS SLIMTX1MIX_Input_3_ W_STS SLIMTX1MIX_SRC3[7:0] 0000h R1988 SLIMTX1MIX_Input_3_ W_STS SLIMTX1MIX_SRC3[7:0] 0000h 0 0 0 0 0 0000h R1988 SLIMTX1MIX_Input_3_ U_STS SLIMTX1MIX_SRC3[7:0] 00000h 0 0 0 0000h R1988 SLIMTX1MIX_Input_4_ W_STS SLIMTX1MIX_SRC3[7:0] 0	R1985	SLIMTX1MIX_Input_1_	-	0	0	0	0	0	0	0	SLIMTX1MIX_VOL1 [6:0] 0	0080h
Closely Description <	R1986	SLIMTX1MIX_Input_2_		0	0	0	0	0	0	0	SLIMTX1MIX_SRC2 [7:0]	0000h
R1986 SLIMTX1MIX_Input_3_ SLIMTX1MIX_STG 0	R1987	SLIMTX1MIX_Input_2_	-	0	0	0	0	0	0	0	SLIMTX1MIX_VOL2 [6:0] 0	0080h
(10,17) DOIS D 0	R1988			0	0	0	0	0	0	0	SLIMTX1MIX_SRC3 [7:0]	0000h
R1990 SLIMTX1MIX_Input_4_ SLIMTX1MIX_Input_4_ SLIMTX1MIX_Input_4_ 0	()		-	0	0	0	0	0	0	0	SLIMTX1MIX_VOL3 [6:0] 0	0080h
(7C6h) Source IX_STS IX_STS <thix_sts< th=""> <thix_sts< th=""> <thix_sts< td="" th<=""><td></td><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>SLIMTX1MIX_SRC4 [7:0]</td><td>0000h</td></thix_sts<></thix_sts<></thix_sts<>				0	0	0	0	0	0	0	SLIMTX1MIX_SRC4 [7:0]	0000h
(7C7h) Volume - <th< td=""><td></td><td>Source</td><td>-</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>SLIMTX1MIX VOL4 [6:0] 0</td><td>0080h</td></th<>		Source	-	0	0	0	0	0	0	0	SLIMTX1MIX VOL4 [6:0] 0	0080h
(7C8h) Source IX_STS IX_STS IX_STS IX_STS IX_STS IX_STS IX_STS IX_STS 0	(7C7h)	Volume	SLIMTX2M	0	0	0	0	0	0	0		
(7C9h) Volume - <th< td=""><td>(7C8h)</td><td>Source</td><td>IX_STS</td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td></th<>	(7C8h)	Source	IX_STS									
(7CAh) Source IX_STS IX_STS <thix_sts< th=""> <thix_sts< th=""> <thix_sts< td="" th<=""><td>(7C9h)</td><td>Volume</td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td></thix_sts<></thix_sts<></thix_sts<>	(7C9h)	Volume										
(7CBh) Volume - <th< td=""><td>(7CAh)</td><td>Source</td><td>IX_STS</td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td></th<>	(7CAh)	Source	IX_STS									
(7CCh) Source IX_STS I <thi< th=""> I I</thi<>	(7CBh)	Volume										
(7CDh) Volume - <th< td=""><td>(7CCh)</td><td>Source</td><td>IX_STS</td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td></th<>	(7CCh)	Source	IX_STS									
(7CEh) Source V = 1 IX_STS	(7CDh)	Volume										
(7CFh) Volume - <th< td=""><td>(7CEh)</td><td>Source</td><td>IX_STS</td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td><td></td></th<>	(7CEh)	Source	IX_STS									
(7D0h) Source IX_STS IIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIIII	(7CFh)	Volume										
	(7D0h)	Source	IX_STS									0000h
(nTU) volume	R2001 (7D1h)	SLIMTX3MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	SLIMTX3MIX_VOL1 [6:0] 0	0080h



Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
R2002 (7D2h)	SLIMTX3MIX_Input_2_ Source	SLIMTX3M IX STS	0	0	0	0	0	0	0	SLIMTX3MIX_SRC2 [7:0]	0000h
(7D21) R2003 (7D3h)	SLIMTX3MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	SLIMTX3MIX_VOL2 [6:0] 0	0080h
R2004 (7D4h)	SLIMTX3MIX_Input_3_ Source	SLIMTX3M IX_STS	0	0	0	0	0	0	0	SLIMTX3MIX_SRC3 [7:0]	0000h
R2005 (7D5h)	SLIMTX3MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	SLIMTX3MIX_VOL3 [6:0] 0	0080h
R2006 (7D6h)	SLIMTX3MIX_Input_4_ Source	SLIMTX3M IX_STS	0	0	0	0	0	0	0	SLIMTX3MIX_SRC4 [7:0]	0000h
R2007 (7D7h)	SLIMTX3MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	SLIMTX3MIX_VOL4 [6:0] 0	0080h
R2008 (7D8h)	SLIMTX4MIX_Input_1_ Source	SLIMTX4M IX_STS	0	0	0	0	0	0	0	SLIMTX4MIX_SRC1 [7:0]	0000h
R2009 (7D9h)	SLIMTX4MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	SLIMTX4MIX_VOL1 [6:0] 0	0080h
R2010 (7DAh)	SLIMTX4MIX_Input_2_ Source	SLIMTX4M IX_STS	0	0	0	0	0	0	0	SLIMTX4MIX_SRC2 [7:0]	0000h
R2011 (7DBh)	SLIMTX4MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	SLIMTX4MIX_VOL2 [6:0] 0	0080h
R2012 (7DCh)	SLIMTX4MIX_Input_3_ Source	SLIMTX4M IX_STS	0	0	0	0	0	0	0	SLIMTX4MIX_SRC3 [7:0]	0000h
R2013 (7DDh)	SLIMTX4MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	SLIMTX4MIX_VOL3 [6:0] 0	0080h
R2014 (7DEh)	SLIMTX4MIX_Input_4_ Source	SLIMTX4M IX_STS	0	0	0	0	0	0	0	SLIMTX4MIX_SRC4 [7:0]	0000h
R2015 (7DFh)	SLIMTX4MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	SLIMTX4MIX_VOL4 [6:0] 0	0080h
R2016 (7E0h)	SLIMTX5MIX_Input_1_ Source	SLIMTX5M IX_STS	0	0	0	0	0	0	0	SLIMTX5MIX_SRC1 [7:0]	0000h
R2017 (7E1h)	SLIMTX5MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	SLIMTX5MIX_VOL1 [6:0] 0	0080h
R2018 (7E2h)	SLIMTX5MIX_Input_2_ Source	SLIMTX5M IX_STS	0	0	0	0	0	0	0	SLIMTX5MIX_SRC2 [7:0]	0000h
R2019 (7E3h)	SLIMTX5MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	SLIMTX5MIX_VOL2 [6:0] 0	0080h
R2020 (7E4h)	SLIMTX5MIX_Input_3_ Source	SLIMTX5M IX_STS	0	0	0	0	0	0	0	SLIMTX5MIX_SRC3 [7:0]	0000h
R2021 (7E5h)	SLIMTX5MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	SLIMTX5MIX_VOL3 [6:0] 0	0080h
R2022 (7E6h)	SLIMTX5MIX_Input_4_ Source	SLIMTX5M IX_STS	0	0	0	0	0	0	0	SLIMTX5MIX_SRC4 [7:0]	0000h
R2023 (7E7h)	SLIMTX5MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	SLIMTX5MIX_VOL4 [6:0] 0	0080h
R2024 (7E8h)	SLIMTX6MIX_Input_1_ Source	SLIMTX6M IX_STS	0	0	0	0	0	0	0	SLIMTX6MIX_SRC1 [7:0]	0000h
R2025 (7E9h)	SLIMTX6MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	SLIMTX6MIX_VOL1 [6:0] 0	0080h
R2026 (7EAh)	SLIMTX6MIX_Input_2_ Source	SLIMTX6M IX_STS	0	0	0	0	0	0	0	SLIMTX6MIX_SRC2 [7:0]	0000h
R2027 (7EBh)	SLIMTX6MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	SLIMTX6MIX_VOL2 [6:0] 0	0080h
R2028 (7ECh)	SLIMTX6MIX_Input_3_ Source	SLIMTX6M IX_STS	0	0	0	0	0	0	0	SLIMTX6MIX_SRC3 [7:0]	0000h
R2029 (7EDh)	SLIMTX6MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	SLIMTX6MIX_VOL3 [6:0] 0	0080h
R2030 (7EEh)	SLIMTX6MIX_Input_4_ Source	SLIMTX6M IX_STS	0	0	0	0	0	0	0	SLIMTX6MIX_SRC4 [7:0]	0000h
R2031 (7EFh)	SLIMTX6MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	SLIMTX6MIX_VOL4 [6:0] 0	0080h
R2032 (7F0h)	SLIMTX7MIX_Input_1_ Source	SLIMTX7M IX_STS	0	0	0	0	0	0	0	SLIMTX7MIX_SRC1 [7:0]	0000h
R2033 (7F1h)	SLIMTX7MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	SLIMTX7MIX_VOL1 [6:0] 0	0080h
R2034 (7F2h)	SLIMTX7MIX_Input_2_ Source	SLIMTX7M IX_STS	0	0	0	0	0	0	0	SLIMTX7MIX_SRC2 [7:0]	0000h
R2035 (7F3h)	SLIMTX7MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	SLIMTX7MIX_VOL2 [6:0] 0	0080h
R2036 (7F4h)	SLIMTX7MIX_Input_3_ Source	SLIMTX7M IX_STS	0	0	0	0	0	0	0	SLIMTX7MIX_SRC3 [7:0]	0000h
R2037 (7F5h)	SLIMTX7MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	SLIMTX7MIX_VOL3 [6:0] 0	0080h
R2038 (7F6h)	SLIMTX7MIX_Input_4_ Source	SLIMTX7M IX_STS	0	0	0	0	0	0	0	SLIMTX7MIX_SRC4 [7:0]	0000h
R2039 (7F7h)	SLIMTX7MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	SLIMTX7MIX_VOL4 [6:0] 0	0080h



(FFB) Boards (FFB) (FFB) Boards (FFB)	Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
(FFB) Nume <	R2040 (7F8h)			0	0	0	0	0	0	0	SLIMTX8MIX_SRC1 [7:0]	0000h
(iFA) Bauco (iFA) C <thc< th=""> C C C <</thc<>		Volume	0	0	0	0	0	0	0	0		0080h
(FRD) Nume Num Nume Nume <t< td=""><td></td><td>SLIMTX8MIX_Input_2_ Source</td><td>SLIMTX8M IX_STS</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>SLIMTX8MIX_SRC2 [7:0]</td><td>0000h</td></t<>		SLIMTX8MIX_Input_2_ Source	SLIMTX8M IX_STS	0	0	0	0	0	0	0	SLIMTX8MIX_SRC2 [7:0]	0000h
(FFC) Burne P< P< <td></td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>SLIMTX8MIX_VOL2 [6:0] 0</td> <td>0080h</td>			0	0	0	0	0	0	0	0	SLIMTX8MIX_VOL2 [6:0] 0	0080h
(FFD) Nume <			SLIMTX8M IX_STS	0	0	0	0	0	0	0	SLIMTX8MIX_SRC3 [7:0]	0000h
OFED: Bauros C <thc< th=""> C <thc< td=""><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>SLIMTX8MIX_VOL3 [6:0] 0</td><td>0080h</td></thc<></thc<>			0	0	0	0	0	0	0	0	SLIMTX8MIX_VOL3 [6:0] 0	0080h
(FFF) Nohme POPTI 0 0			SLIMTX8M IX_STS	0	0	0	0	0	0	0	SLIMTX8MIX_SRC4 [7:0]	0000h
(a) (a) (a) (a) (a) (a) (a) (b) (b) (c) (b) (c) (c) (c) (c) (c) (c) (c) (c) (c) (c			0	0	0	0	0	0	0	0	SLIMTX8MIX_VOL4 [6:0] 0	0080h
(a) Unive Lower Lower <thlower< th=""> Lower Lower</thlower<>				0	0	0	0	0	0	0	SPDIF1TX1_SRC [7:0]	0000h
(800) (Source - <t< td=""><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>SPDIF1TX1_VOL [6:0] 0</td><td>0080h</td></t<>			0	0	0	0	0	0	0	0	SPDIF1TX1_VOL [6:0] 0	0080h
(a) (a) (b) (c) (c) <td></td> <td></td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>SPDIF1TX2_SRC [7:0]</td> <td>0000h</td>				0	0	0	0	0	0	0	SPDIF1TX2_SRC [7:0]	0000h
(B8DH) Source Image: Simple interval			0	0	0	0	0	0	0	0	SPDIF1TX2_VOL [6:0] 0	0080h
(8811) Volume Image: Constraint of the second sec				0	0	0	0	0	0	0	EQ1MIX_SRC1 [7:0]	0000h
Barrie Source ST Image: Source String Control EXTME			0	0	0	0	0	0	0	0	EQ1MIX_VOL1 [6:0] 0	0080h
(8837) Volume (1) (0	0	0	0	0	0	0	EQ1MIX_SRC2 [7:0]	0000h
(B6M) Source STS Image: Source STS Image: Source			0	0	0	0	0	0	0	0	EQ1MIX_VOL2 [6:0] 0	0080h
(885) Molume - - - -			EQ1MIX_ STS	0	0	0	0	0	0	0	EQ1MIX_SRC3 [7:0]	0000h
(886) Source ST I <t< td=""><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>EQ1MIX_VOL3 [6:0] 0</td><td>0080h</td></t<>			0	0	0	0	0	0	0	0	EQ1MIX_VOL3 [6:0] 0	0080h
(87) Volume C			EQ1MIX_ STS	0	0	0	0	0	0	0	EQ1MIX_SRC4 [7:0]	0000h
(888) Source STS Image: Constraint of the state			0	0	0	0	0	0	0	0	EQ1MIX_VOL4 [6:0] 0	0080h
(889) Volume C <thc< td=""><td></td><td></td><td>EQ2MIX_ STS</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>EQ2MIX_SRC1 [7:0]</td><td>0000h</td></thc<>			EQ2MIX_ STS	0	0	0	0	0	0	0	EQ2MIX_SRC1 [7:0]	0000h
(i88Ah) Source F.T. O		EQ2MIX_Input_1_ Volume	0	0	0	0	0	0	0	0	EQ2MIX_VOL1 [6:0] 0	0080h
(88h) Volume Volume Volume Volume 0<				0	0	0	0	0	0	0	EQ2MIX_SRC2 [7:0]	0000h
(860h) Source STS Image: Constraint of the state of the sta			0	0	0	0	0	0	0	0	EQ2MIX_VOL2 [6:0] 0	0080h
(88Dh) Molume r <th< td=""><td></td><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>EQ2MIX_SRC3 [7:0]</td><td>0000h</td></th<>				0	0	0	0	0	0	0	EQ2MIX_SRC3 [7:0]	0000h
Cost /r Cost /r <t< td=""><td></td><td></td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>0</td><td>EQ2MIX_VOL3 [6:0] 0</td><td>0080h</td></t<>			0	0	0	0	0	0	0	0	EQ2MIX_VOL3 [6:0] 0	0080h
(88Fh) Volume Volume<			EQ2MIX_ STS	0	0	0	0	0	0	0	EQ2MIX_SRC4 [7:0]	0000h
(890h) Source T STS			0	0	0	0	0	0	0	0	EQ2MIX_VOL4 [6:0] 0	0080h
R2193 (891h) EQ3MIX_Input_1_ Volume 0	R2192			0	0	0	0	0	0	0	EQ3MIX_SRC1 [7:0]	0000h
R2194 EQ3MIX_Input_2_ Source EQ3MIX_STS 0 0 0 0 0 0 EQ3MIX_SRC2 [7:0] 0000h R2195 EQ3MIX_Input_2_ Volume 0 0 0 0 0 0 0 0 0 0 0 0000h R2195 EQ3MIX_Input_2_ Volume 0	R2193	EQ3MIX_Input_1_	0	0	0	0	0	0	0	0	EQ3MIX_VOL1 [6:0] 0	0080h
R2195 EQ3MIX_Input_2_ 0	R2194	EQ3MIX_Input_2_	EQ3MIX_ STS	0	0	0	0	0	0	0	EQ3MIX_SRC2 [7:0]	0000h
R2196 (894h) EQ3MIX_Input_3_ Source EQ3MIX_STS 0	R2195	EQ3MIX_Input_2_	0	0	0	0	0	0	0	0	EQ3MIX_VOL2 [6:0] 0	0080h
R2197 EQ3MIX_Input_3_ Volume 0 </td <td>R2196</td> <td>EQ3MIX_Input_3_ Source</td> <td>EQ3MIX_ STS</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>EQ3MIX_SRC3 [7:0]</td> <td>0000h</td>	R2196	EQ3MIX_Input_3_ Source	EQ3MIX_ STS	0	0	0	0	0	0	0	EQ3MIX_SRC3 [7:0]	0000h
R2198 EQ3MIX_Input_4_ Source EQ3MIX_STS 0	R2197		0	0	0	0	0	0	0	0	EQ3MIX_VOL3 [6:0] 0	0080h
R2199 (897h) EQ3MIX_Input_4_ Volume 0	R2198		EQ3MIX_ STS	0	0	0	0	0	0	0	EQ3MIX_SRC4 [7:0]	0000h
(898h) Source STS STS STS State State <thstate< th=""> State Stat</thstate<>	R2199	EQ3MIX_Input_4_	0	0	0	0	0	0	0	0	EQ3MIX_VOL4 [6:0] 0	0080h
R2201 EQ4MIX_Input_1_ 0 0 0 0 0 0 0 0 0080h				0	0	0	0	0	0	0	EQ4MIX_SRC1 [7:0]	0000h
			0	0	0	0	0	0	0	0	EQ4MIX_VOL1 [6:0] 0	0080h



Register	Name	15	14	13	12	11	10	9	8	7 6 5 4 3 2 1 0	Default
R2202 (89Ah)	EQ4MIX_Input_2_ Source	EQ4MIX_ STS	0	0	0	0	0	0	0	EQ4MIX_SRC2 [7:0]	0000h
R2203 (89Bh)	EQ4MIX_Input_2_ Volume	0	0	0	0	0	0	0	0	EQ4MIX_VOL2 [6:0] 0	0080h
R2204 (89Ch)	EQ4MIX_Input_3_ Source	EQ4MIX_ STS	0	0	0	0	0	0	0	EQ4MIX_SRC3 [7:0]	0000h
R2205 (89Dh)	EQ4MIX_Input_3_ Volume	0	0	0	0	0	0	0	0	EQ4MIX_VOL3 [6:0] 0	0080h
R2206 (89Eh)	EQ4MIX_Input_4_ Source	EQ4MIX_ STS	0	0	0	0	0	0	0	EQ4MIX_SRC4 [7:0]	0000h
R2207 (89Fh)	EQ4MIX_Input_4_ Volume	0	0	0	0	0	0	0	0	EQ4MIX_VOL4 [6:0] 0	0080h
R2240 (8C0h)	DRC1LMIX_Input_1_ Source	DRC1LMIX _STS	0	0	0	0	0	0	0	DRC1LMIX_SRC1 [7:0]	0000h
R2241 (8C1h)	DRC1LMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	DRC1LMIX_VOL1 [6:0] 0	0080h
R2242 (8C2h)	DRC1LMIX_Input_2_ Source	DRC1LMIX _STS	0	0	0	0	0	0	0	DRC1LMIX_SRC2 [7:0]	0000h
R2243 (8C3h)	DRC1LMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	DRC1LMIX_VOL2 [6:0] 0	0080h
R2244 (8C4h)	DRC1LMIX_Input_3_ Source	DRC1LMIX _STS	0	0	0	0	0	0	0	DRC1LMIX_SRC3 [7:0]	0000h
R2245 (8C5h)	DRC1LMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	DRC1LMIX_VOL3 [6:0] 0	0080h
R2246 (8C6h)	DRC1LMIX_Input_4_ Source	DRC1LMIX _STS	0	0	0	0	0	0	0	DRC1LMIX_SRC4 [7:0]	0000h
R2247 (8C7h)	DRC1LMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	DRC1LMIX_VOL4 [6:0] 0	0080h
R2248 (8C8h)	DRC1RMIX_Input_1_ Source	DRC1RMI X_STS	0	0	0	0	0	0	0	DRC1RMIX_SRC1 [7:0]	0000h
R2249 (8C9h)	DRC1RMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	DRC1RMIX_VOL1 [6:0] 0	0080h
R2250 (8CAh)	DRC1RMIX_Input_2_ Source	DRC1RMI X_STS	0	0	0	0	0	0	0	DRC1RMIX_SRC2 [7:0]	0000h
R2251 (8CBh)	DRC1RMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	DRC1RMIX_VOL2 [6:0] 0	0080h
R2252 (8CCh)	DRC1RMIX_Input_3_ Source	DRC1RMI X_STS	0	0	0	0	0	0	0	DRC1RMIX_SRC3 [7:0]	0000h
R2253 (8CDh)	DRC1RMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	DRC1RMIX_VOL3 [6:0] 0	0080h
R2254 (8CEh)	DRC1RMIX_Input_4_ Source	DRC1RMI X_STS	0	0	0	0	0	0	0	DRC1RMIX_SRC4 [7:0]	0000h
R2255 (8CFh)	DRC1RMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	DRC1RMIX_VOL4 [6:0] 0	0080h
R2256 (8D0h)	DRC2LMIX_Input_1_ Source	DRC2LMIX _STS	0	0	0	0	0	0	0	DRC2LMIX_SRC1 [7:0]	0000h
R2257 (8D1h)	DRC2LMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	DRC2LMIX_VOL1 [6:0] 0	0080h
R2258 (8D2h)	DRC2LMIX_Input_2_ Source	DRC2LMIX _STS	0	0	0	0	0	0	0	DRC2LMIX_SRC2 [7:0]	0000h
R2259 (8D3h)	DRC2LMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	DRC2LMIX_VOL2 [6:0] 0	0080h
R2260 (8D4h)	DRC2LMIX_Input_3_ Source	DRC2LMIX _STS	0	0	0	0	0	0	0	DRC2LMIX_SRC3 [7:0]	0000h
R2261 (8D5h)	DRC2LMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	DRC2LMIX_VOL3 [6:0] 0	0080h
R2262 (8D6h)	DRC2LMIX_Input_4_ Source	DRC2LMIX _STS	0	0	0	0	0	0	0	DRC2LMIX_SRC4 [7:0]	0000h
R2263 (8D7h)	DRC2LMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	DRC2LMIX_VOL4 [6:0] 0	0080h
R2264 (8D8h)	DRC2RMIX_Input_1_ Source	DRC2RMI X_STS	0	0	0	0	0	0	0	DRC2RMIX_SRC1 [7:0]	0000h
R2265 (8D9h)	DRC2RMIX_Input_1_ Volume	0	0	0	0	0	0	0	0	DRC2RMIX_VOL1 [6:0] 0	0080h
R2266 (8DAh)	DRC2RMIX_Input_2_ Source	DRC2RMI X_STS	0	0	0	0	0	0	0	DRC2RMIX_SRC2 [7:0]	0000h
R2267 (8DBh)	DRC2RMIX_Input_2_ Volume	0	0	0	0	0	0	0	0	DRC2RMIX_VOL2 [6:0] 0	0080h
R2268 (8DCh)	DRC2RMIX_Input_3_ Source	DRC2RMI X_STS	0	0	0	0	0	0	0	DRC2RMIX_SRC3 [7:0]	0000h
R2269 (8DDh)	DRC2RMIX_Input_3_ Volume	0	0	0	0	0	0	0	0	DRC2RMIX_VOL3 [6:0] 0	0080h
R2270 (8DEh)	DRC2RMIX_Input_4_ Source	DRC2RMI X_STS	0	0	0	0	0	0	0	DRC2RMIX_SRC4 [7:0]	0000h
R2271 (8DFh)	DRC2RMIX_Input_4_ Volume	0	0	0	0	0	0	0	0	DRC2RMIX_VOL4 [6:0] 0	0080h



(900h) S R2305 H (901h) W R2306 H (902h) S R2307 H (903h) W R2308 H	HPLP1MIX_Input_1_ Source HPLP1MIX_Input_1_ /olume HPLP1MIX_Input_2_ Source	LHPF1MIX _STS 0	0	0	0	0	0	0	0	LHPF1MIX SRC1 [7:0]	0000h
(901h) M R2306 H (902h) S R2307 H (903h) M R2308 H	/olume IPLP1MIX_Input_2_	0							ů		000011
(902h) S R2307 H (903h) V R2308 H			0	0	0	0	0	0	0	LHPF1MIX_VOL1 [6:0] 0	0080h
(903h) V R2308 H		LHPF1MIX _STS	0	0	0	0	0	0	0	LHPF1MIX_SRC2 [7:0]	0000h
	IPLP1MIX_Input_2_ /olume	0	0	0	0	0	0	0	0	LHPF1MIX_VOL2 [6:0] 0	0080h
(904h) S	HPLP1MIX_Input_3_ Source	LHPF1MIX _STS	0	0	0	0	0	0	0	LHPF1MIX_SRC3 [7:0]	0000h
R2309 H (905h) V	IPLP1MIX_Input_3_ /olume	0	0	0	0	0	0	0	0	LHPF1MIX_VOL3 [6:0] 0	0080h
	HPLP1MIX_Input_4_ Source	LHPF1MIX _STS	0	0	0	0	0	0	0	LHPF1MIX_SRC4 [7:0]	0000h
	IPLP1MIX_Input_4_ /olume	0	0	0	0	0	0	0	0	LHPF1MIX_VOL4 [6:0] 0	0080h
	HPLP2MIX_Input_1_ Source	LHPF2MIX _STS	0	0	0	0	0	0	0	LHPF2MIX_SRC1 [7:0]	0000h
(909h) V	IPLP2MIX_Input_1_ /olume	0	0	0	0	0	0	0	0	LHPF2MIX_VOL1 [6:0] 0	0080h
(90Ah) S	HPLP2MIX_Input_2_ Source	LHPF2MIX _STS	0	0	0	0	0	0	0	LHPF2MIX_SRC2 [7:0]	0000h
(90Bh) V	IPLP2MIX_Input_2_ /olume	0	0	0	0	0	0	0	0	LHPF2MIX_VOL2 [6:0] 0	0080h
(90Ch) S	HPLP2MIX_Input_3_ Source	LHPF2MIX _STS	0	0	0	0	0	0	0	LHPF2MIX_SRC3 [7:0]	0000h
	IPLP2MIX_Input_3_ /olume	0	0	0	0	0	0	0	0	LHPF2MIX_VOL3 [6:0] 0	0080h
(90Eh) S	HPLP2MIX_Input_4_ Source	LHPF2MIX _STS	0	0	0	0	0	0	0	LHPF2MIX_SRC4 [7:0]	0000h
	IPLP2MIX_Input_4_ /olume	0	0	0	0	0	0	0	0	LHPF2MIX_VOL4 [6:0] 0	0080h
(910h) S	HPLP3MIX_Input_1_ Source	LHPF3MIX _STS	0	0	0	0	0	0	0	LHPF3MIX_SRC1 [7:0]	0000h
(911h) V	HPLP3MIX_Input_1_ /olume	0	0	0	0	0	0	0	0	LHPF3MIX_VOL1 [6:0] 0	0080h
	HPLP3MIX_Input_2_ Source	LHPF3MIX _STS	0	0	0	0	0	0	0	LHPF3MIX_SRC2 [7:0]	0000h
	IPLP3MIX_Input_2_ /olume	0	0	0	0	0	0	0	0	LHPF3MIX_VOL2 [6:0] 0	0080h
(914h) S	HPLP3MIX_Input_3_ Source	LHPF3MIX _STS	0	0	0	0	0	0	0	LHPF3MIX_SRC3 [7:0]	0000h
(915h) V	IPLP3MIX_Input_3_ /olume	0	0	0	0	0	0	0	0	LHPF3MIX_VOL3 [6:0] 0	0080h
(916h) S	HPLP3MIX_Input_4_ Source	LHPF3MIX _STS	0	0	0	0	0	0	0	LHPF3MIX_SRC4 [7:0]	0000h
(917h) V	IPLP3MIX_Input_4_ /olume	0	0	0	0	0	0	0	0	LHPF3MIX_VOL4 [6:0] 0	0080h
(918h) S	HPLP4MIX_Input_1_ Source	LHPF4MIX _STS	0	0	0	0	0	0	0	LHPF4MIX_SRC1 [7:0]	0000h
(919h) V	HPLP4MIX_Input_1_ /olume	0	0	0	0	0	0	0	0	LHPF4MIX_VOL1 [6:0] 0	0080h
(91Ah) S	HPLP4MIX_Input_2_ Source	LHPF4MIX _STS	0	0	0	0	0	0	0	LHPF4MIX_SRC2 [7:0]	0000h
(91Bh) V	IPLP4MIX_Input_2_ /olume	0	0	0	0	0	0	0	0	LHPF4MIX_VOL2 [6:0] 0	0080h
	HPLP4MIX_Input_3_ Source	LHPF4MIX _STS	0	0	0	0	0	0	0	LHPF4MIX_SRC3 [7:0]	0000h
(91Dh) V	IPLP4MIX_Input_3_ /olume	0	0	0	0	0	0	0	0	LHPF4MIX_VOL3 [6:0] 0	0080h
	HPLP4MIX_Input_4_ Source	LHPF4MIX _STS	0	0	0	0	0	0	0	LHPF4MIX_SRC4 [7:0]	0000h
(91Fh) V	IPLP4MIX_Input_4_ /olume	0	0	0	0	0	0	0	0	LHPF4MIX_VOL4 [6:0] 0	0080h
(940h) S	DSP1LMIX_Input_1_ Source	DSP1LMIX _STS	0	0	0	0	0	0	0	DSP1LMIX_SRC1 [7:0]	0000h
(941h) V	OSP1LMIX_Input_1_ /olume	0	0	0	0	0	0	0	0	DSP1LMIX_VOL1 [6:0] 0	0080h
(942h) S	DSP1LMIX_Input_2_ Source	DSP1LMIX _STS	0	0	0	0	0	0	0	DSP1LMIX_SRC2 [7:0]	0000h
(943h) V	OSP1LMIX_Input_2_ /olume	0	0	0	0	0	0	0	0	DSP1LMIX_VOL2 [6:0] 0	0080h
(944h) S	DSP1LMIX_Input_3_ Source	DSP1LMIX _STS	0	0	0	0	0	0	0	DSP1LMIX_SRC3 [7:0]	0000h
	OSP1LMIX_Input_3_ /olume	0	0	0	0	0	0	0	0	DSP1LMIX_VOL3 [6:0] 0	0080h



Register Name 15 14 13 12 11 10 9 R2374 DSP1LMIX_Input_4_ DSP1LMIX_SUP DSP1LMIX_STS 0	8 7 6 5 4 3 2 1 0 0 DSP1LMIX_SRC4 [7:0] DSP1LMIX_SRC4 [7:0] 0	Default 0000h 0080h 0000h 0080h 0000h 0080h 0000h 0080h 0000h 0080h 0000h 0080h 0080h
R2375 DSP1LMIX_Input_4_ 0	0 DSP1RMIX_SRC1 [7:0] 0 DSP1RMIX_VOL1 [6:0] 0 0 DSP1RMIX_SRC2 [7:0] 0 0 DSP1RMIX_VOL2 [6:0] 0 0 DSP1RMIX_SRC3 [7:0] 0 0 DSP1RMIX_VOL3 [6:0] 0	0000h 0080h 0000h 0080h 0080h
R2376 DSP1RMIX_Input_1_ DSP1RMI 0<	0 DSP1RMIX_VOL1 [6:0] 0 0 DSP1RMIX_SRC2 [7:0] 0 0 DSP1RMIX_VOL2 [6:0] 0 0 DSP1RMIX_SRC3 [7:0] 0 0 DSP1RMIX_VOL3 [6:0] 0	0080h 0000h 0080h 0000h
R2377 DSP1RMIX_Input_1_ 0	0 DSP1RMIX_SRC2 [7:0] 0 DSP1RMIX_VOL2 [6:0] 0 0 DSP1RMIX_SRC3 [7:0] 0 0 DSP1RMIX_VOL3 [6:0] 0	0000h 0080h 0000h
R2378 DSP1RMIX_Input_2_ DSP1RMI 0<	0 DSP1RMIX_VOL2 [6:0] 0 0 DSP1RMIX_SRC3 [7:0] 0 0 DSP1RMIX_VOL3 [6:0] 0	0080h 0000h
R2379 DSP1RMIX_Input_2_ 0	0 DSP1RMIX_SRC3 [7:0] 0 DSP1RMIX_VOL3 [6:0] 0	0000h
R2380 DSP1RMIX_Input_3_ DSP1RMI 0<	0 DSP1RMIX_VOL3 [6:0] 0	
(94Dh) Volume R2382 DSP1RMIX_Input_4_ DSP1RMI 0 0 0 0 0		0080h
	0 DSP1RMIX_SRC4 [7:0]	
		0000h
R2383 DSP1RMIX_Input_4_ 0	0 DSP1RMIX_VOL4 [6:0] 0	0080h
R2384 DSP1AUX1MIX_Input_ DSP1AUX 0	0 DSP1AUX1_SRC [7:0]	0000h
R2392 DSP1AUX2MIX_Input_ DSP1AUX 0	0 DSP1AUX2_SRC [7:0]	0000h
R2400 DSP1AUX3MIX_Input_ DSP1AUX 0	0 DSP1AUX3_SRC [7:0]	0000h
R2408 DSP1AUX4MIX_Input_ DSP1AUX 0	0 DSP1AUX4_SRC [7:0]	0000h
R2416 DSP1AUX5MIX_Input_ DSP1AUX 0	0 DSP1AUX5_SRC [7:0]	0000h
R2424 DSP1AUX6MIX_Input_ DSP1AUX 0	0 DSP1AUX6_SRC [7:0]	0000h
R2688 ASRC1_1LMIX_Input_ ASRC1_0 0	0 ASRC1_IN1L_SRC [7:0]	0000h
R2696 ASRC1_1RMIX_Input_ ASRC1_0 0	0 ASRC1_IN1R_SRC [7:0]	0000h
R2704 ASRC1_2LMIX_Input_ (A90h) ASRC1_0 0	0 ASRC1_IN2L_SRC [7:0]	0000h
R2712 ASRC1_2RMIX_Input_ ASRC1_0 0	0 ASRC1_IN2R_SRC [7:0]	0000h
R2816 ISRC1DEC1MIX_Input_ ISRC1DEC 0 <th< td=""><td>0 ISRC1DEC1_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC1DEC1_SRC [7:0]	0000h
R2824 ISRC1DEC2MIX_Input_ ISRC1DEC 0 <th< td=""><td>0 ISRC1DEC2_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC1DEC2_SRC [7:0]	0000h
R2848 ISRC1INT1MIX_Input_ ISRC1INT 0 <th< td=""><td>0 ISRC1INT1_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC1INT1_SRC [7:0]	0000h
R2856 ISRC1INT2MIX_Input_ ISRC1INT 0 <th< td=""><td>0 ISRC1INT2_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC1INT2_SRC [7:0]	0000h
R2880 ISRC2DEC1MIX_Input_ ISRC2DEC 0 <th< td=""><td>0 ISRC2DEC1_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC2DEC1_SRC [7:0]	0000h
R2888 ISRC2DEC2MIX_Input_ ISRC2DEC_0 <	0 ISRC2DEC2_SRC [7:0]	0000h
R2912 ISRC2INT1MIX_Input_ ISRC2INT 0 <th< td=""><td>0 ISRC2INT1_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC2INT1_SRC [7:0]	0000h
R2920 ISRC2INT2MIX_Input_ ISRC2INT 0 <th< td=""><td>0 ISRC2INT2_SRC [7:0]</td><td>0000h</td></th<>	0 ISRC2INT2_SRC [7:0]	0000h
R3520 DFC1MIX_Input_1_ DFC1_ 0 <td>0 DFC1_SRC [7:0]</td> <td>0000h</td>	0 DFC1_SRC [7:0]	0000h
R3528 DFC2MIX_Input_1_ DFC2_STS 0<	0 DFC2_SRC [7:0]	0000h
R3536 DFC3MIX_Input_1_ DFC3_ 0 <td>0 DFC3_SRC [7:0]</td> <td>0000h</td>	0 DFC3_SRC [7:0]	0000h
R3544 DFC4MIX_Input_1_ DFC4_STS_ 0	0 DFC4_SRC [7:0]	0000h
R3552 (DE0h) DFC5MIX_Input_1_ DFC5_ STS 0	0 DFC5_SRC [7:0]	0000h
R3560 DFC6MIX_Input_1_ DFC6_STS 0<	0 DFC6_SRC [7:0]	0000h
R3568 DFC7MIX_Input_1_ DFC7_STS 0<	0 DFC7_SRC [7:0]	0000h
R3576 DFC8MIX_Input_1_ DFC8_ 0 <td>0 DFC8_SRC [7:0]</td> <td>0000h</td>	0 DFC8_SRC [7:0]	0000h
R3584 FX_Ctrl1 FX_RATE [4:0] 0 0 (E00h)	0 0 0 0 0 0 0 0 0	0000h
R3585 FX_Ctrl2 FX_STS [11:0] (E01h)	0 0 1 0	0002h



Record Field	Register	Name	15 14 13 12 11	10 9 8 7 6	5	4	3	2	1	0	Default
R888 F01.2 F01.9_L0A/L0A/L0A C C C C C D <thd< th=""> D D <thd< th=""></thd<></thd<>		EQ1_1	EQ1_B1_GAIN [4:0]	EQ1_B2_GAIN [4:0]		EQ	1_B3_GAII	N [4:0]		EQ1_ENA	6318h
(cf:0) 01.4 COTE 0 [15] 0150 00560 8008 801.5 601.5 6050 6050 8008 801.6 601.7 6050 6050 6050 8008 801.7 601.9 600.9 <td>R3601</td> <td>EQ1_2</td> <td>EQ1_B4_GAIN [4:0]</td> <td></td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>0</td> <td>EQ1_B1_ MODE</td> <td>6300h</td>	R3601	EQ1_2	EQ1_B4_GAIN [4:0]		0	0	0	0	0	EQ1_B1_ MODE	6300h
Image:		EQ1_3		EQ1_B1_A [15:0]							0FC8h
Refer Corr 0.000, A (56) FECA A REMER Corr 1.000, FECA A FECA A REMER Corr 1.0000, FECA A FECA A	(E13h)			EQ1_B1_B [15:0]							03FEh
(affin) (b) (c)	R3604 (E14h)	EQ1_5		EQ1_B1_PG [15:0]							00E0h
(if-th) Contract of the Chief	R3605 (E15h)	EQ1_6		EQ1_B2_A [15:0]							1EC4h
Image: Bit in the set of the set	(E16h)										F136h
(fellen) COGBN COGBN <thcodsn< th=""> COGBN COGBN <</thcodsn<>	(E17h)			EQ1_B2_C [15:0]							0409h
image: constraint of the	R3608 (E18h)	EQ1_9		EQ1_B2_PG [15:0]							04CCh
(fit Ref 1)		EQ1_10		EQ1_B3_A [15:0]							1C9Bh
CETCH COCCBA COCBA COCBA <thcocba< th=""> <thcocba< th=""> <thcocba< th=""></thcocba<></thcocba<></thcocba<>	R3610 (E1Ah)	EQ1_11		EQ1_B3_B [15:0]							F337h
CETCO CO CO <thc< td=""><td></td><td>EQ1_12</td><td></td><td>EQ1_B3_C [15:0]</td><td></td><td></td><td></td><td></td><td></td><td></td><td>040Bh</td></thc<>		EQ1_12		EQ1_B3_C [15:0]							040Bh
GETOR Control F709n R844 G115 G116 F709n G40Ah R845 G116 G117 G114,JPG [159] G40Ah R846 G117 G114,JPG [159] G505 G505 R847 G0118 G0119 G118,JPG [159] G505 G505 R247 G0118 G0119 G0119,SP150 G505 G605 R2481 G0120 E019,SP150 G505 G505 G505 R2481 G0120 E012,SP150 G505 G505 <td< td=""><td>R3612</td><td>EQ1_13</td><td></td><td>EQ1_B3_PG [15:0]</td><td></td><td></td><td></td><td></td><td></td><td></td><td>0CBBh</td></td<>	R3612	EQ1_13		EQ1_B3_PG [15:0]							0CBBh
Image: Control (Control (Contro) (Contro) (Control (Contro) (Contro) (Contro) (Contro) (Contro) (EQ1_14		EQ1_B4_A [15:0]							16F8h
Image: definition		EQ1_15		EQ1_B4_B [15:0]							F7D9h
(ECh) (ECh) (GR) <		EQ1_16		EQ1_B4_C [15:0]							040Ah
Image: Control of the contro	R3616 (E20h)	EQ1_17		EQ1_B4_PG [15:0]							1F14h
Image: Constraint of the	R3617 (E21h)	EQ1_18		EQ1_B5_A [15:0]							058Ch
Image: Constraint of the		EQ1_19		EQ1_B5_B [15:0]							0563h
(E2Ah) (E2Bh) (E02_B1_GAN [4:0]) E02_B1_GAN [4:0] (E02_B2_GAN [4:0]) (E02_B3_GAN [4:0]) (E02_B1_GAN [4:0]) <	R3619 (E23h)	EQ1_20		EQ1_B5_PG [15:0]							4000h
(E26h) - C <td></td> <td>EQ1_21</td> <td></td> <td>EQ1_B1_C [15:0]</td> <td></td> <td></td> <td></td> <td></td> <td></td> <td></td> <td>0B75h</td>		EQ1_21		EQ1_B1_C [15:0]							0B75h
(EZ7h) C (MODE) R3624 EQ2_3 EQ2_4 0FC8h (E28h) EQ2_4 EQ2_B1_R[150] 0FC8h R3625 EQ2_4 EQ2_B1_R[150] 00Ebh R3626 EQ2_5 EQ2_6 EQ2_B1_PG[150] 00Ebh R3627 EQ2_6 EQ2_6 EQ2_B1_R[150] 1EC4h R3628 EQ2_7 EQ2_82_R[150] 1EC4h 1EC4h R3628 EQ2_7 EQ2_8_R[150] 1EC4h 1EC4h R3628 EQ2_7 EQ2_8_R[150] 1EC4h 1EC4h R3628 EQ2_7 EQ2_8_R[150] 1EC4h 1EC4h R3628 EQ2_9 EQ2_8_R[150] 1EC4h 1EC4h R3638 EQ2_11 EQ2_8_R[150] 1C9Bh 1C9Bh R3633 EQ2_11 EQ2_8_R[150] F337h 1C9Bh R3633 EQ2_11 EQ2_8_R[150] F337h 1C9Bh R3633 EQ2_12 EQ2_8_R[150] EQ2_8_R[150] 1C9Bh R3634 <t< td=""><td></td><td>EQ2_1</td><td></td><td>EQ2_B2_GAIN [4:0]</td><td></td><td>EQ2</td><td>2_B3_GAII</td><td>N [4:0]</td><td></td><td>EQ2_ENA</td><td>6318h</td></t<>		EQ2_1		EQ2_B2_GAIN [4:0]		EQ2	2_B3_GAII	N [4:0]		EQ2_ENA	6318h
[E28h] C C R3625 EQ2_4 EQ2_B1_B(150) Q3FEh R3626 EQ2_5 EQ2_6 00E0h R3627 EQ2_6 EQ2_B1_PG(150) 1EC4h R3627 EQ2_6 EQ2_B2_A(150) 1EC4h R3628 EQ2_7 EQ2_6 F136h R3628 EQ2_7 EQ2_8_B(150) 0409h R3628 EQ2_8 EQ2_9 0409h R3630 EQ2_9 EQ2_9 0409h R3631 EQ2_10 EQ2_83_A(150) 04CCh (E2Ph) EQ2_11 EQ2_9 EQ2_83_B(150) 1C9Bh (E30h) EQ2_11 EQ2_83_B(150) 602Bh 1C9Bh (E30h) EQ2_11 EQ2_83_B(150) 602Bh 1C9Bh (E30h) EQ2_11 EQ2_83_B(150) 1C9Bh 1C9Bh (E30h) EQ2_11 EQ2_83_B(150) 1C9Bh 1C9Bh (E30h) EQ2_11 EQ2_84_A(150) 1C9Bh 1C9Bh (E33h)		EQ2_2	EQ2_B4_GAIN [4:0]	EQ2_B5_GAIN [4:0]	0	0	0	0	0	EQ2_B1_ MODE	6300h
(E29h) COUNT COUNT <t< td=""><td></td><td>EQ2_3</td><td></td><td>EQ2_B1_A [15:0]</td><td></td><td></td><td></td><td></td><td></td><td></td><td>0FC8h</td></t<>		EQ2_3		EQ2_B1_A [15:0]							0FC8h
(E2A) - <td></td> <td>EQ2_4</td> <td></td> <td>EQ2_B1_B [15:0]</td> <td></td> <td></td> <td></td> <td></td> <td></td> <td></td> <td>03FEh</td>		EQ2_4		EQ2_B1_B [15:0]							03FEh
(E2Bh) C C R3628 EQ2_7 EQ2_B2_B[150] F136h R3629 EQ2_8 EQ2_B2_C[150] 0409h (E2Dh) EQ2_9 EQ2_910 040Ch R3631 EQ2_10 EQ2_B3_C[150] 1C9Bh (E2Fh) EQ2_11 EQ2_B3_C[150] 1C9Bh R3633 EQ2_12 EQ2_B3_C[150] 1C9Bh (E3th) EQ2_13 EQ2_B3_C[150] 040Bh R3635 EQ2_13 EQ2_B3_C[150] 040Bh (E33h) EQ2_14 EQ2_B3_PG[150] 040Bh (E33h) EQ2_12 EQ2_B3_PG[150] 040Bh (E33h) EQ2_13 EQ2_14 0CBBh (E33h) EQ2_14 EQ2_B3_PG[150] 0CBBh (E33h) EQ2_14 EQ2_B4_F[150] 0CBBh (E33h) EQ2_15 EQ2_B4_F[150] 6CBA (E33h) EQ2_16 EQ2_B4_F[150] 6Q4Ah (E33h) EQ2_16 EQ2_B4_F[150] 1F14h		EQ2_5		EQ2_B1_PG [15:0]							00E0h
$\begin{array}{ c c c c } \hline (c2Ch) & \hline (c2Ch$	R3627 (E2Bh)	EQ2_6		EQ2_B2_A [15:0]							1EC4h
(E2Dh) - <td>R3628 (E2Ch)</td> <td>EQ2_7</td> <td></td> <td>EQ2_B2_B [15:0]</td> <td></td> <td></td> <td></td> <td></td> <td></td> <td></td> <td>F136h</td>	R3628 (E2Ch)	EQ2_7		EQ2_B2_B [15:0]							F136h
R3630 (E2Eh) EQ2_9 EQ2_9(15.0) 04CCh R3631 (E2Fh) EQ2_10 EQ2_B3_A[15.0] 1C9Bh R3632 (E3Ch) EQ2_11 EQ2_B3_B[15.0] F337h R3633 (E31h) EQ2_12 EQ2_B3_C[15.0] 040Bh R3633 (E31h) EQ2_13 040Bh 040Bh R3634 (E32h) EQ2_14 EQ2_B3_PG[15.0] 040Bh R3635 (E33h) EQ2_14 EQ2_B1_PG[15.0] 040Bh R3636 (E33h) EQ2_14 EQ2_B4_A[15.0] 16F8h R3636 (E33h) EQ2_15 EQ2_16 EQ2_B4_C[15.0] 040Ah R3637 (E33h) EQ2_16 EQ2_B4_PG[15.0] 1714h	R3629	EQ2_8		EQ2_B2_C [15:0]							0409h
R3631 (E2Fh) EQ2_10 EQ2_B3_A[15:0] 1C9Bh R3632 (E30h) EQ2_11 EQ2_B3_B[15:0] F337h R3633 EQ2_12 EQ2_B3_C[15:0] 040Bh R3634 (E31h) EQ2_13 EQ2_B3_C[15:0] 0CBBh R3635 (E32h) EQ2_14 EQ2_B4_R[15:0] 0CBBh R3636 (E33h) EQ2_15 EQ2_B4_R[15:0] 16F8h R3636 (E33h) EQ2_16 EQ2_B4_C[15:0] 040Ah R3637 EQ2_16 EQ2_B4_C[15:0] 040Ah	R3630	EQ2_9		EQ2_B2_PG [15:0]							04CCh
R3632 (E30h) EQ2_11 EQ2_B3_B[15:0] F337h R3633 (E31h) EQ2_12 EQ2_B3_C[15:0] 040Bh R3634 (E32h) EQ2_13 EQ2_B3_FG[15:0] 040Bh R3635 (E33h) EQ2_14 0CBBh 0CBBh R3636 (E33h) EQ2_15 16F8h 16F8h R3636 (E33h) EQ2_15 EQ2_B4_B[15:0] F7D9h R3637 (E33h) EQ2_16 EQ2_B4_C[15:0] 040Ah R3638 EQ2_17 EQ2_B4_PG[15:0] 1F14h	R3631	EQ2_10		EQ2_B3_A [15:0]							1C9Bh
R3633 (E31h) EQ2_12 EQ2_B3_C [15:0] 040Bh R3634 (E32h) EQ2_13 EQ2_B3_PG [15:0] 0CBBh R3635 (E32h) EQ2_14 EQ2_B4_A [15:0] 16F8h R3636 (E33h) EQ2_15 EQ2_B4_B [15:0] F7D9h R3637 (E33h) EQ2_16 EQ2_B4_C [15:0] 040Ah R3638 (E33h) EQ2_17 EQ2_B4_C [15:0] 1F14h	R3632	EQ2_11		EQ2_B3_B [15:0]							F337h
R3634 (E32h) EQ2_13 EQ2_13 OCBBh R3635 (E33h) EQ2_14 EQ2_B4_A [15:0] 16F8h R3636 (E33h) EQ2_15 EQ2_B4_B [15:0] 17D9h R3637 (E34h) EQ2_16 EQ2_B4_C [15:0] 040Ah R3638 EQ2_17 EQ2_B4_C [15:0] 1F14h	R3633	EQ2_12		EQ2_B3_C [15:0]							040Bh
R3635 (E33h) EQ2_14 EQ2_B4_A[15:0] 16F8h R3636 (E33h) EQ2_15 EQ2_B4_B[15:0] F7D9h R3637 (E33h) EQ2_16 EQ2_B4_C[15:0] 040Ah R3638 EQ2_17 EQ2_B4_PG[15:0] 1F14h	R3634	EQ2_13		EQ2_B3_PG [15:0]							0CBBh
R3636 (E34h) EQ2_15 EQ2_B4_B [15:0] F7D9h R3637 (E35h) EQ2_16 EQ2_B4_C [15:0] 040Ah R3638 EQ2_17 EQ2_B4_PG [15:0] 1F14h	R3635	EQ2_14		EQ2_B4_A [15:0]							16F8h
R3637 (E35h) EQ2_16 EQ2_B4_C [15:0] 040Ah R3638 EQ2_17 EQ2_B4_C [15:0] 1F14h	R3636	EQ2_15		EQ2_B4_B [15:0]							F7D9h
R3638 EQ2_17 EQ2_B4_PG [15:0] 1F14h	R3637	EQ2_16		EQ2_B4_C [15:0]							040Ah
		EQ2_17		EQ2_B4_PG [15:0]							1F14h



Register	Name	15 14 13 12 11	10 9 8 7 6	5	4	3	2	1	0	Default					
R3639 (E37h)	EQ2_18		EQ2_B5_A [15:0]							058Ch					
R3640 (E38h)	EQ2_19		EQ2_B5_B [15:0]							0563h					
	EQ2_20		EQ2_B5_PG [15:0]							4000h					
	EQ2_21		EQ2_B1_C [15:0]							0B75h					
	EQ3_1	EQ3_B1_GAIN [4:0]	EQ3_B2_GAIN [4:0]		EQ	B3_GAIN	N [4:0]		EQ3_ENA	6318h					
	EQ3_2	EQ3_B4_GAIN [4:0]	EQ3_B5_GAIN [4:0]	0	0	0	0	0	EQ3_B1_ MODE	6300h					
	EQ3_3		EQ3_B1_A [15:0]	11		1	I	1		0FC8h					
	EQ3_4		EQ3_B1_B [15:0]							03FEh					
R3648 (E40h)	EQ3_5		EQ3_B1_PG [15:0]							00E0h					
R3649 (E41h)	EQ3_6		EQ3_B2_A [15:0]							1EC4h					
R3650 (E42h)	EQ3_7		EQ3_B2_B [15:0]							F136h					
R3651 (E43h)	EQ3_8		EQ3_B2_C [15:0]							0409h					
(E44h)	EQ3_9		EQ3_B2_PG [15:0]							04CCh					
R3653 (E45h)	EQ3_10		EQ3_B3_A [15:0]							1C9Bh					
R3654 (E46h)	EQ3_11		EQ3_B3_B [15:0]							F337h					
R3655 (E47h)	EQ3_12		EQ3_B3_C [15:0]							040Bh					
R3656 (E48h)	EQ3_13		EQ3_B3_PG [15:0]							0CBBh					
(E49h)	EQ3_14		EQ3_B4_A [15:0]												
R3658 (E4Ah)	EQ3_15	EQ3_B4_B [15:0] F													
R3659 (E4Bh)	EQ3_16		EQ3_B4_C [15:0]												
R3660 (E4Ch)	EQ3_17		EQ3_B4_PG [15:0]							1F14h					
R3661 (E4Dh)	EQ3_18		EQ3_B5_A [15:0]							058Ch					
(E4Eh)	EQ3_19		EQ3_B5_B [15:0]							0563h					
(E4Fh)	EQ3_20		EQ3_B5_PG [15:0]							4000h					
(E50h)	EQ3_21		EQ3_B1_C [15:0]							0B75h					
(E52h)	EQ4_1	EQ4_B1_GAIN [4:0]	EQ4_B2_GAIN [4:0]		EQ4	LB3_GAIN	N [4:0]		EQ4_ENA	6318h					
(E53h)	EQ4_2	EQ4_B4_GAIN [4:0]	EQ4_B5_GAIN [4:0]	0	0	0	0	0	EQ4_B1_ MODE	6300h					
(E54h)	EQ4_3		EQ4_B1_A [15:0]							0FC8h					
R3669 (E55h)	EQ4_4		EQ4_B1_B [15:0]							03FEh					
(E56h)	EQ4_5		EQ4_B1_PG [15:0]							00E0h					
(E57h)	EQ4_6		EQ4_B2_A [15:0]							1EC4h					
R3672 (E58h)	EQ4_7		EQ4_B2_B [15:0]							F136h					
R3673 (E59h)	EQ4_8		EQ4_B2_C [15:0]							0409h					
R3674 (E5Ah)	EQ4_9		EQ4_B2_PG [15:0]							04CCh					
R3675 (E5Bh)	EQ4_10		EQ4_B3_A [15:0]							1C9Bh					
	EQ4_11		EQ4_B3_B [15:0]							F337h					
R3677 (E5Dh)	EQ4_12		EQ4_B3_C [15:0]							040Bh					
	EQ4_13		EQ4_B3_PG [15:0]							0CBBh					



Register	Naille	19	14	15	12		10	3	0	1	0	Ð	4	3	2	I	U	Delault
R3679 (E5Fh)	EQ4_14								EQ4_B4	_A [15:0]								16F8h
R3680 (E60h)	EQ4_15								EQ4_B4	_B [15:0]								F7D9h
R3681 (E61h)	EQ4_16								EQ4_B4	_C [15:0]								040Ah
R3682 (E62h)	EQ4_17								EQ4_B4_	_PG [15:0]								1F14h
R3683 (E63h)	EQ4_18								EQ4_B5	_A [15:0]								058Ch
R3684 (E64h)	EQ4_19								EQ4_B5	_B [15:0]								0563h
R3685 (E65h)	EQ4_20								EQ4_B5_	_PG [15:0]								4000h
R3686 (E66h)	EQ4_21								EQ4_B1	_C [15:0]								0B75h
R3712 (E80h)	DRC1_ctrl1		DRC1_S	SIG_DET_F	RMS [4:0]		DRC1_SIC [1	G_DET_PK :0]	DRC1_ NG_ENA	DRC1_ SIG_DET_ MODE	DRC1_ SIG_DET	DRC1_ KNEE2_ OP_ENA	DRC1_QF	R DRC1 ANTICLĪP	DRC1_ WSEQ_ SIG_DET_ ENA	DRC1L_ ENA	DRC1R_ ENA	0018h
R3713 (E81h)	DRC1_ctrl2	0	0	0		DRC1_	ATK [3:0]			DRC1_E	DCY [3:0]		DRC	C1_MINGAIN		DRC1_MA	XGAIN [1:0]	0933h
R3714 (E82h)	DRC1_ctrl3	[DRC1_NG_N	/INGAIN [3	:0]	DRC1_NG	6_EXP [1:0]	DRC1_QR		DRC1_QR	_DCY [1:0]	DRC	1_HI_COM	P [2:0]	DRC1	LO_COM	P [2:0]	0018h
R3715 (E83h)	DRC1_ctrl4	0	0	0	0	0			DRC1_KN	EE_IP [5:0]				DRC	1_KNEE_O	P [4:0]		0000h
R3716 (E84h)	DRC1_ctrl5	0	0	0	0	0	0		DRC	1_KNEE2_I	P [4:0]			DRC1	_KNEE2_C)P [4:0]		0000h
	DRC2_ctrl1		DRC2_S	SIG_DET_F	MS [4:0]			G_DET_PK :0]	DRC2_ NG_ENA	DRC2_ SIG_DET_ MODE	DRC2_ SIG_DET	DRC2_ KNEE2_ OP_ENA	DRC2_QF	R DRC2 ANTICLIP	0	DRC2L_ ENA	DRC2R_ ENA	0018h
R3721 (E89h)	DRC2_ctrl2	0	0	0		DRC2_/	ATK [3:0]			DRC2_E	DCY [3:0]		DRC	2_MINGAI	N [2:0]	DRC2_MA	XGAIN [1:0]	0933h
R3722 (E8Ah)	DRC2_ctrl3	0	DRC2_NG_N	AINGAIN [3	:0]	DRC2_NG	G_EXP [1:0]	DRC2_QR	_THR [1:0]	DRC2_QR	L_DCY [1:0]	DRC	2_HI_COM	P [2:0]	DRC2	2_LO_COM	P [2:0]	0018h
R3723 (E8Bh)	DRC2_ctrl4	0	0	0	0	0		1	DRC2_KN	EE_IP [5:0]				DRC	2_KNEE_O	P [4:0]		0000h
R3724 (E8Ch)	DRC2_ctrl5	0	0	0	0	0	0		DRC2	2_KNEE2_I	P [4:0]			DRC2	2_KNEE2_C	OP [4:0]		0000h
R3776 (EC0h)	HPLPF1_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	LHPF1_ MODE	LHPF1_ ENA	0000h
R3777 (EC1h)	HPLPF1_2				1			1	LHPF1_CC	DEFF [15:0]								0000h
R3780 (EC4h)	HPLPF2_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	LHPF2_ MODE	LHPF2_ ENA	0000h
R3781 (EC5h)	HPLPF2_2				1			1	LHPF2_CO	DEFF [15:0]								0000h
R3784 (EC8h)	HPLPF3_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	LHPF3_ MODE	LHPF3_ ENA	0000h
R3785 (EC9h)	HPLPF3_2				1			1	LHPF3_CO	DEFF [15:0]								0000h
R3788 (ECCh)	HPLPF4_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	LHPF4_ MODE	LHPF4_ ENA	0000h
R3789 (ECDh)	HPLPF4_2			ı		ı	ı	ı	LHPF4_CC	DEFF [15:0]	1	ı	ı		<u>، </u>	ı		0000h
R3808 (EE0h)	ASRC1_ENABLE	0	0	0	0	0	0	0	0	0	0	0	0	ASRC1_ IN2L_ENA	ASRC1_ IN2R_ENA	ASRC1_ IN1L_ENA	ASRC1_ IN1R_ENA	0000h
R3809 (EE1h)	ASRC1_STATUS	0	0	0	0	0	0	0	0	0	0	0	0	ASRC1_ IN2L_	ASRC1_ IN2R_	ASRC1_ IN1L_	ASRC1_ IN1R_ ENA_STS	0000h
R3810 (EE2h)	ASRC1_RATE1		ASR	C1_RATE1	[4:0]	1	0	0	0	0	0	0	0	ENA_SIS 0	ENA_STS 0	ENA_SIS 0	ENA_SIS 0	0000h
R3811 (EE3h)	ASRC1_RATE2		ASR	C1_RATE2	[4:0]		0	0	0	0	0	0	0	0	0	0	0	4000h
R3824 (EF0h)	ISRC1_CTRL_1		ISI	RC1_FSH [4	4:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R3825 (EF1h)	ISRC1_CTRL_2		ISI	RC1_FSL [4	4:0]		0	0	0	0	0	0	0	0	0	0	1	0001h
R3826 (EF2h)	ISRC1_CTRL_3	ISRC1_ INT1_ENA	ISRC1_ INT2_ENA	0	0	0	0	ISRC1_ DEC1_ ENA	ISRC1_ DEC2_ ENA	0	0	0	0	0	0	0	0	0000h
R3827 (EF3h)	ISRC2_CTRL_1		ISF	RC2_FSH [4	4:0]	1	0	0	0	0	0	0	0	0	0	0	0	0000h
R3828 (EF4h)	ISRC2_CTRL_2		ISI	RC2_FSL [4	4:0]		0	0	0	0	0	0	0	0	0	0	1	0001h
R3829 (EF5h)	ISRC2_CTRL_3	ISRC2_ INT1_ENA	ISRC2_ AINT2_ENA	0	0	0	0	ISRC2_ DEC1_ ENA	ISRC2_ DEC2_ ENA	0	0	0	0	0	0	0	0	0000h
R4224 (1080h)	US1_Ctrl_0	0	0	US1_G	AIN [1:0]		US1_S	RC [3:0]	•	0	U	S1_FREQ [2	2:0]	0	0	0	US1_ENA	2030h



Name

Register

Table 6-1. Register Map Definition—16-bit region (Cont.)

15 14 13 12 11 10 9 8 7 6 5 4 3 2 1 0 Default



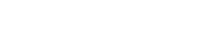
Table 6-1.	Register Map Definition—16-bit region (Cont.)	

Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R4225 (1081h)	US1_Ctrl_1		US	61_RATE [4	:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R4226 (1082h)	US2_Ctrl_0	0	0	US2_G/	AIN [1:0]		US2_SI	RC [3:0]		0	US	52_FREQ [2	2:0]	0	0	0	US2_ENA	2030h
R4227 (1083h)	US2_Ctrl_1		US	62_RATE [4	:0]		0	0	0	0	0	0	0	0	0	0	0	0000h
R4288 (10C0h)	AUXPDM1_Ctrl_0	0	0	0	0		AUXPDM1	I_SRC[3:0]		0	0	0	AUXPDM1 _TXEDGE	AUXPDM1 _MSTR	AUXPDM1 _MUTE	0	AUXPDM1 _ENA	0008h
R4289 (10C1h)	AUXPDM1_Ctrl_1	AUXPDM FREG	/1_CLK_ Q[1:0]	0	0	0	0	0	0	0	0	0	0	0	0	0	0	4000h
R5248 (1480h)	DFC1_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C1_RATE [4	4:0]		DFC1_ DITH_ENA		0000h
R5250 (1482h)	DFC1_RX_W0	0	0	0		_	K_DATA_WI			0	0	0	0	0		X_DATA_T		1F00h
R5252 (1484h)	DFC1_TX_W0	0	0	0		DFC1_T>	(_DATA_WI	DTH [4:0]		0	0	0	0	0	DFC1_T.	X_DATA_T	YPE [2:0]	1F00h
R5254 (1486h)	DFC2_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C2_RATE [4	4:0]		DFC2_ DITH_ENA	DFC2_ ENA	0000h
R5256 (1488h)	DFC2_RX_W0	0	0	0		_	K_DATA_WI			0	0	0	0	0	DFC2_R	X_DATA_T	YPE [2:0]	1F00h
R5258 (148Ah)	DFC2_TX_W0	0	0	0		DFC2_T>	(_DATA_WI	DTH [4:0]		0	0	0	0	0	DFC2_T	X_DATA_T		1F00h
R5260 (148Ch)	DFC3_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C3_RATE [4	4:0]		DFC3_ DITH_ENA	DFC3_ ENA	0000h
R5262 (148Eh)	DFC3_RX_W0	0	0	0			K_DATA_WI			0	0	0	0	0		X_DATA_T		1F00h
R5264 (1490h)	DFC3_TX_W0	0	0	0		DFC3_T>	(_DATA_WI	DTH [4:0]		0	0	0	0	0	DFC3_T.	X_DATA_T	YPE [2:0]	1F00h
R5266 (1492h)	DFC4_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C4_RATE [4	4:0]		DFC4_ DITH_ENA		0000h
R5268 (1494h)	DFC4_RX_W0	0	0	0		DFC4_R)	K_DATA_WI	IDTH [4:0]		0	0	0	0	0	DFC4_R	X_DATA_T	YPE [2:0]	1F00h
R5270 (1496h)	DFC4_TX_W0	0	0	0		DFC4_T>	(_DATA_WI	DTH [4:0]		0	0	0	0	0	DFC4_T.	X_DATA_T	YPE [2:0]	1F00h
R5272 (1498h)	DFC5_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C5_RATE [4	4:0]		DFC5_ DITH_ENA	DFC5_ ENA	0000h
R5274 (149Ah)	DFC5_RX_W0	0	0	0		DFC5_R	K_DATA_WI	IDTH [4:0]		0	0	0	0	0	DFC5_R	X_DATA_T	YPE [2:0]	1F00h
R5276 (149Ch)	DFC5_TX_W0	0	0	0		DFC5_T>	K_DATA_WI	DTH [4:0]		0	0	0	0	0	DFC5_T.	X_DATA_T	YPE [2:0]	1F00h
R5278 (149Eh)	DFC6_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C6_RATE [4	4:0]		DFC6_ DITH_ENA	DFC6_ ENA	0000h
R5280 (14A0h)	DFC6_RX_W0	0	0	0		DFC6_R>	K_DATA_WI	IDTH [4:0]		0	0	0	0	0		X_DATA_T		1F00h
R5282 (14A2h)	DFC6_TX_W0	0	0	0		DFC6_T>	(_DATA_WI	DTH [4:0]		0	0	0	0	0	DFC6_T.	X_DATA_T		1F00h
R5284 (14A4h)	DFC7_CTRL_W0	0	0	0	0	0	0	0	0	0			C7_RATE [4			DFC7_ DITH_ENA	DFC7_ ENA	0000h
R5286 (14A6h)	DFC7_RX_W0	0	0	0			K_DATA_WI			0	0	0	0	0	_	X_DATA_T		1F00h
R5288 (14A8h)	DFC7_TX_W0	0	0	0			(_DATA_WI			0	0	0	0	0	DFC7_T.	X_DATA_T		1F00h
R5290 (14AAh)	DFC8_CTRL_W0	0	0	0	0	0	0	0	0	0		DF	C8_RATE [4	4:0]		DFC8_ DITH_ENA	DFC8_ ENA	0000h
R5292 (14ACh)	DFC8_RX_W0	0	0	0		-	(_DATA_WI			0	0	0	0	0		X_DATA_T		1F00h
R5294 (14AEh)	DFC8_TX_W0	0	0	0		-	(_DATA_WI		-	0	0	0	0	0	-	X_DATA_T	YPE [2:0]	1F00h
R5302 (14B6h)	DFC_STATUS_W0	0	0	0	0	0	0	0	0		T			_CHAN [7:0]				0000h
R5303 (14B7h)	DFC_STATUS_W1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	- [1	[H_TYPE :0]	0000h
R5632 (1600h)	ADSP2_IRQ0	0	0	0	0	0	0	0	0	0	0	0	0	0		DSP_IRQ2		0000h
(1601h)	ADSP2_IRQ1	0	0	0	0	0	0	0	0	0	0	0	0	0		DSP_IRQ4		0000h
R5634 (1602h)	ADSP2_IRQ2	0	0	0	0	0	0	0	0	0	0	0	0	0		DSP_IRQ6		0000h
R5635 (1603h)	ADSP2_IRQ3	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP_IRQ8		0000h
R5636 (1604h)	ADSP2_IRQ4	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP IRQ10	DSP_IRQ9	0000h
R5637 (1605h)	ADSP2_IRQ5	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP IRQ12	DSP IRQ11	0000h
R5638 (1606h)	ADSP2_IRQ6	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP IRQ14	DSP IRQ13	0000h



Register R5639	Name	15	14															
	ADSP2 IRQ7	0	0	13 0	12 0	11 0	10	9 0	8 0	7 0	6 0	5 0	4 0	3 0	2 0	1 DSP	0 DSP	Default 0000h
(1607h) R5888	GPIO1 CTRL 1	GP1 LVL	GP1 OP	GP1 DB	GP1 POL	0	0	Ů	Ů	Ŭ	Ŭ		N [9:0]	Ū	Ů	IRQ16	IRQ15	2001h
(1700h) R5889	GPIO1_CTRL_2	- GP1_DIR	CFG - GP1_PU	- GP1_PD	_ GP1_	0	0	0	0	0	0	0	0	0	0	0	0	F000h
(1701h) R5890	GPIO2_CTRL_1	GP2_LVL	GP2_OP_ CFG	GP2_DB	DRV_STR GP2_POL	0	0					GP2_F	N [9:0]					2001h
(1702h) R5891	GPIO2_CTRL_2	GP2_DIR	CFG GP2_PU	GP2_PD	GP2_	0	0	0	0	0	0	0	0	0	0	0	0	F000h
(1703h) R5892	GPIO3_CTRL_1	GP3_LVL	GP3_OP_	GP3_DB	DRV_STR GP3_POL	0	0					GP3_F	N [9:0]					2001h
(1704h) R5893	GPIO3_CTRL_2	GP3_DIR	CFG GP3_PU	GP3_PD	GP3_	0	0	0	0	0	0	0	0	0	0	0	0	F000h
(1705h) R5894	GPIO4_CTRL_1	GP4_LVL	GP4_OP_ CFG	GP4_DB	DRV_STR GP4_POL	0	0					GP4_F	N [9:0]					2001h
(1706h) R5895	GPIO4_CTRL_2	GP4_DIR	GP4_PU	GP4_PD	GP4_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
(1707h) R5896	GPIO5_CTRL_1	GP5_LVL	GP5_OP_ CFG	GP5_DB	GP5_POL	0	0					GP5_F	N [9:0]					2001h
(1708h) R5897 (1709h)	GPIO5_CTRL_2	GP5_DIR	GP5_PU	GP5_PD	GP5_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5898 (170Ah)	GPIO6_CTRL_1	GP6_LVL	GP6_OP_ CFG	GP6_DB	- GP6_POL	0	0					GP6_F	N [9:0]				1	2001h
	GPIO6_CTRL_2	GP6_DIR	GP6_PU	GP6_PD	GP6_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5900 (170Ch)	GPIO7_CTRL_1	GP7_LVL	GP7_OP_ CFG	GP7_DB	GP7_POL	0	0			1		GP7_F	N [9:0]				1	2001h
R5901 (170Dh)	GPIO7_CTRL_2	GP7_DIR	GP7_PU	GP7_PD	GP7_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5902 (170Eh)	GPIO8_CTRL_1	GP8_LVL	GP8_OP_ CFG	GP8_DB	GP8_POL	0	0			1		GP8_F	N [9:0]				1	2001h
R5903 (170Fh)	GPIO8_CTRL_2	GP8_DIR	GP8_PU	GP8_PD	GP8_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5904 (1710h)	GPIO9_CTRL_1	GP9_LVL	GP9_OP_ CFG	GP9_DB	GP9_POL	0	0		•	•		GP9_F	N [9:0]		•		•	2001h
R5905 (1711h)	GPIO9_CTRL_2	GP9_DIR	GP9_PU	GP9_PD	GP9_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5906 (1712h)	GPIO10_CTRL_1	GP10_LVL	GP10_ OP_CFG	GP10_DB	GP10_ POL	0	0					GP10_	FN [9:0]					2001h
R5907 (1713h)	GPIO10_CTRL_2	_	GP10_PU	GP10_PD	GP10_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5908 (1714h)	GPIO11_CTRL_1	GP11_LVL	CFG	-	GP11_POL	0	0						FN [9:0]					2001h
R5909 (1715h)	GPIO11_CTRL_2	GP11_DIR	-	GP11_PD	GP11_ DRV_STR	0	0	0	0	0	0	0	0	0	0	0	0	F000h
R5910 (1716h)	GPIO12_CTRL_1	GP12_LVL	GP12_ OP_CFG	GP12_DB	GP12_ POL	0	0						FN [9:0]					2001h
R5911 (1717h)	GPIO12_CTRL_2	_	GP12_PU	GP12_PD	GP12_ DRV_STR		0	0	0	0	0	0	0	0	0	0	0	F000h
(1718h)	GPIO13_CTRL_1	GP13_LVL	OP_CFG	GP13_DB	POL	0	0		0				FN [9:0]	0			<u>^</u>	2001h
(1719h)	GPIO13_CTRL_2	_	GP13_PU	-	GP13_ DRV_STR	0	0	0	0	0	0	0	0 FN [9:0]	0	0	0	0	F000h
R5914 (171Ah) R5915	GPIO14_CTRL_1 GPIO14 CTRL 2	GP14_LVL	OP_CFG	GP14_DB GP14 PD	GP14_ POL GP14	0	0	0	0	0	0	GP14_	РМ [9:0] 0	0	0	0	0	2001h F000h
(171Bh)	GPI014_CTRL_2	GP14_DIK	_	GP14_PD GP15 DB	DRV_STR GP15	0	0	U	U	U	U		FN [9:0]	Ū	U	U	0	2001h
(171Ch)	GPIO15_CTRL_1	_	OP_CFG GP15 PU	-	POL GP15_	0	0	0	0	0	0	0	0	0	0	0	0	F000h
(171Dh)	GPIO16 CTRL 1	GP16 LVL	_	GP16 DB	DRV_STR GP16		0		Ů	Ŭ	Ŭ		FN [9:0]	Ŭ	Ů	Ŭ	Ů	2001h
(171Eh) R5919	GPIO16 CTRL 2	_	OP_CFG GP16 PU	_	POL GP16	0	0	0	0	0	0	0	0	0	0	0	0	F000h
(171Fh) R6144	IRQ1 Status 1	0	0	0	DRV_STR	0	0	SYSCLK_	0	BOOT_	0	0	0	0	0	0	0	0000h
(1800h)					ERR_ EINT1			FAIL EINT1		DONE_ EINT1								
R6145 (1801h)	IRQ1_Status_2	0	0	CLK ASYNC_ ERR_ EINT1	CLK_SYS_ ERR_ EINT1	0	0	FLL2_ LOCK_ EINT1	FLL1_ LOCK_ EINT1	0	FLL2_ REF_ LOST_ EINT1	FLL1_ REF_ LOST_ EINT1	0	0	0	0	0	0000h
R6149 (1805h)	IRQ1_Status_6	0	0	0	0	0	0	MICDET2_ EINT1	MICDET1_ EINT1	0	0	0	0	0	0	0	HPDET_ EINT1	0000h
R6150 (1806h)	IRQ1_Status_7	0	0	0	0	MICD_ CLAMP2_ FALL_ EINT1	MICD_ CLAMP2_ RISE_ EINT1	JD3_ FALL_ EINT1	JD3 RISE EINT1	0	0	MICD_ CLAMP1_ FALL_ EINT1	MICD CLAMP1_ RISE_ EINT1	JD2_ FALL_ EINT1	JD2 RISE EINT1	JD1_ FALL_ EINT1	JD1_ RISE_ EINT1	0000h

CIRRUS LOGIC[®]





	Та	able 6	-1. Re	gister	[,] Мар	Defini	tion—	-16-bit	t regio	on (Co	nt.)		
45	44	40	40	44	40	•	0	7	6	E	4	•	2

Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R6152 (1808h)	IRQ1_Status_9	0	0	0	0	0	0	ASRC1_ IN2 LOCK_ EINT1	ASRC1_ IN1 LOCK_ EINT1	0	0	0	0	0	INPUTS_ SIG_DET_ EINT1	DRC2_ SIG_DET_ EINT1	DRC1_ SIG_DET_ EINT1	0000h
R6154 (180Ah)	IRQ1_Status_11	DSP IRQ16 EINT1	DSP IRQ15 EINT1	DSP IRQ14 EINT1	DSP IRQ13 EINT1	DSP IRQ12 EINT1	DSP IRQ11 EINT1	DSP IRQ10_ EINT1	DSP_ IRQ9_ EINT1	DSP_ IRQ8_ EINT1	DSP_ IRQ7_ EINT1	DSP_ IRQ6_ EINT1	DSP_ IRQ5_ EINT1	DSP_ IRQ4_ EINT1	DSP_ IRQ3_ EINT1	DSP_ IRQ2_ EINT1	DSP_ IRQ1_ EINT1	0000h
R6155 (180Bh)	IRQ1_Status_12	0	0	0	0	0	0	HP4R_ SC_EINT1	HP4L_SC_ EINT1	0	0	HP3R SC_EINT1	HP3L_SC_ EINT1	HP2R_ SC_EINT1	HP2L_SC_ EINT1	HP1R SC_EINT1	HP1L_SC_ EINT1	0000h
R6156 (180Ch)	IRQ1_Status_13	0	0	0	0	0	0	0	0	0	0	HP3R ENABLE_ DONE_ EINT1	HP3L ENABLE_ DONE_ EINT1	HP2R ENABLE_ DONE_ EINT1	HP2L ENABLE_ DONE_ EINT1	HP1R ENABLE_ DONE_ EINT1	HP1L ENABLE_ DONE_ EINT1	0000h
R6157 (180Dh)	IRQ1_Status_14	0	0	0	0	0	0	0	0	0	0	HP3R_ DISABLE_ DONE_ EINT1	HP3L_ DISABLE_ DONE_ EINT1	HP2R_ DISABLE_ DONE_ EINT1	HP2L_ DISABLE_ DONE_ EINT1	HP1R_ DISABLE_ DONE_ EINT1	HP1L_ DISABLE_ DONE_ EINT1	0000h
R6158 (180Eh)	IRQ1_Status_15	0	0	0	DFC_ SATURAT E_EINT1	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R6160 (1810h)	IRQ1_Status_17	GP16_ EINT1	GP15_ EINT1	GP14_ EINT1	GP13_ EINT1	GP12 EINT1	GP11_ EINT1	GP10_ EINT1	GP9_ EINT1	GP8_ EINT1	GP7_ EINT1	GP6_ EINT1	GP5_ EINT1	GP4_ EINT1	GP3 EINT1	GP2 EINT1	GP1_ EINT1	0000h
R6164 (1814h)	IRQ1_Status_21	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	TIMER1_ EINT1	0000h
R6165 (1815h)	IRQ1_Status_22	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ NOT EMPTY_ EINT1	0000h
R6166 (1816h)	IRQ1_Status_23	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ FULL_ EINT1	0000h
R6167 (1817h)	IRQ1_Status_24	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ WMARK_ EINT1	0000h
R6168 (1818h)	IRQ1_Status_25	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ DMA_ EINT1	0000h
R6170 (181Ah)	IRQ1_Status_27	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1 START1_ EINT1	0000h
R6171 (181Bh)	IRQ1_Status_28	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ START2_ EINT1	0000h
R6173 (181Dh)	IRQ1_Status_30	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUSY_ EINT1	0000h
R6176 (1820h)	IRQ1_Status_33	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUS_ ERR_ EINT1	0000h
R6179 (1823h)	IRQ1_Status_36	0	0	0	0	0	0	0	0	0	0	0	0	TIMER_ ALM1_ CH4_ EINT1	TIMER_ ALM1_ CH3_ EINT1	TIMER_ ALM1_ CH2_ EINT1	TIMER_ ALM1_ CH1_ EINT1	0000h
R6183 (1827h)	IRQ1_Status_40	0	0	0	0	0	0	FLL2 SYNC ACTIVE EINT1	FLL1_ SYNC_ ACTIVE_ EINT1	0	0	0	0	0	0	0	0	0000h
R6208 (1840h)	IRQ1_Mask_1	0	0	0	IM CTRLĪF_ ERR EINTĪ	0	0	IM_ SYSCLK_ FAIL_ EINT1	0	IM_BOOT_ DONE_ EINT1	0	0	0	0	0	0	0	1200h
R6209 (1841h)	IRQ1_Mask_2	0	1	IM_CLK_ ASYNC_ ERR EINT1	IM_CLK_ SYS_ ERR_ EINT1	0	0	IM_FLL2_ LOCK_ EINT1	IM_FLL1_ LOCK_ EINT1	0	IM_FLL2_ REF_ LOST_ EINTT	IM_FLL1_ REF LOST_ EINT1	0	0	0	0	0	77E0h
R6213 (1845h)	IRQ1_Mask_6	0	0	0	0	0	0	IM_ MICDET2_ EINT1	IM_ MICDET1_ EINT1	0	0	0	0	0	0	0	IM HPDET_ EINT1	0301h
R6214 (1846h)	IRQ1_Mask_7	0	0	0	0	IM_MICD_ CLAMP2_ FALL_ EINT1		IM_JD3_ FALL EINT1	IM_JD3_ RISE EINT1	0	0	IM_MICD_ CLAMP1_ FALL_ EINT1	IM_MICD_ CLAMP1_ RISE_ EINT1	IM_JD2_ FALL EINT1	IM_JD2_ RISE_ EINT1	IM_JD1_ FALL EINT1	IM_JD1_ RISE_ EINT1	0F3Fh
R6216 (1848h)	IRQ1_Mask_9	0	0	0	0	0	0	IM ASRC1_ IN2 LOCK EINT1	IM ASRC1_ IN1 LOCK EINT1	0	0	0	1	1	IM INPUTS SIG_DET_ EINT1	IM_DRC2_ SIG_DET_ EINT1	IM_DRC1_ SIG_DET_ EINT1	031Fh
R6218 (184Ah)	IRQ1_Mask_11	IM_DSP_ IRQ16_ EINT1	IM_DSP_ IRQ15_ EINT1	IM_DSP_ IRQ14_ EINT1	IM_DSP_ IRQ13_ EINT1	IM_DSP_ IRQ12_ EINT1	IM_DSP_ IRQ11_ EINT1	IM_DSP_ IRQ10_ EINT1	IM_DSP_ IRQ9 EINT1	IM_DSP_ IRQ8_ EINT1	IM_DSP_ IRQ7_ EINT1	IM_DSP_ IRQ6 EINT1	IM_DSP_ IRQ5_ EINT1	IM_DSP_ IRQ4 EINT1	IM_DSP_ IRQ3_ EINT1	IM_DSP_ IRQ2_ EINT1	IM_DSP_ IRQ1 EINT1	FFFFh
R6219 (184Bh)	IRQ1_Mask_12	0	0	0	0	0	0	IM_HP4R_ SC_EINT1	IM_HP4L_ SC_EINT1	0	0	IM_HP3R_ SC_EINT1	IM_HP3L_ SC_EINT1	IM_HP2R_ SC_EINT1	IM_HP2L_ SC_EINT1	IM_HP1R_ SC_EINT1	IM_HP1L_ SC_EINT1	033Fh
R6220 (184Ch)	IRQ1_Mask_13	0	0	0	0	0	0	0	0	0	0	IM_HP3R_ ENABLE_ DONE_ EINT1	IM_HP3L_ ENABLE_ DONE_ EINT1	IM_HP2R_ ENABLE_ DONE_ EINT1	IM_HP2L_ ENABLE_ DONE_ EINT1	IM_HP1R_ ENABLE_ DONE_ EINT1	IM_HP1L_ ENABLE_ DONE_ EINT1	003Fh
R6221 (184Dh)	IRQ1_Mask_14	0	0	0	0	0	0	0	0	0	0	IM_HP3R_ DISABLE_ DONE_ EINT1	IM_HP3L_ DISABLE_ DONE_ EINT1	IM HP2R	IM_HP2L_ DISABLE_ DONE_ EINT1	IM HP1R	IM_HP1L_ DISABLE_ DONE_ EINT1	003Fh
R6222	IRQ1_Mask_15	0	0	0	IM_DFC_ SATURAT	0	0	0	0	0	0	0	0	0	0	0	0	1000h



Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R6224	IRQ1_Mask_17	IM_GP16_ EINT1		IM_GP14_ EINT1	IM_GP13_ EINT1	IM_GP12_ EINT1	IM_GP11_ EINT1	IM_GP10_ EINT1	IM_GP9_ EINT1	IM_GP8_ EINT1	IM_GP7_ EINT1	IM_GP6_ EINT1	IM_GP5_ EINT1	IM_GP4_ EINT1	IM_GP3_ EINT1	IM_GP2_ EINT1	IM_GP1_ EINT1	FFFFh
(1850h) R6228 (1854h)	IRQ1_Mask_21	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_ TIMER1_ EINT1	0001h
R6229 (1855h)	IRQ1_Mask_22	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EINTT EVENT1_ NOT EMPTY_ EINT1	0001h
R6230 (1856h)	IRQ1_Mask_23	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM EVENT1_ FULL EINT1	0001h
R6231 (1857h)	IRQ1_Mask_24	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_ EVENT1_ WMARK_ EINT1	0001h
R6232 (1858h)	IRQ1_Mask_25	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ DMA EINT1	0001h
R6234 (185Ah)	IRQ1_Mask_27	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ START1_ EINT1	0001h
R6235 (185Bh)	IRQ1_Mask_28	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ START2_ EINT1	0001h
R6237 (185Dh)	IRQ1_Mask_30	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ BUSY_ EINT1	0001h
R6240 (1860h)	IRQ1_Mask_33	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ BUS_ ERR_ EINT1	0001h
R6243 (1863h)	IRQ1_Mask_36	0	0	0	0	0	0	0	0	0	0	0	0	IM TIMER_ ALM1_ CH4 EINT1	IM_ TIMER_ ALM1_ CH3_ EINT1	IM TIMER_ ALM1_ CH2_ EINT1	IM_ TIMER_ ALM1_ CH1_ EINT1	000Fh
R6247 (1867h)	IRQ1_Mask_40	0	0	0	0	0	0	IM_FLL2_ SYNC_ ACTIVE_ EINT1	IM_FLL1_ SYNC_ ACTIVE_ EINT1	0	0	0	0	0	0	0	0	0300h
R6272 (1880h)	IRQ1_Raw_Status_1	0	0	0	CTRLIF_ ERR_ STS1	0	0	0	0	BOOT_ DONE_ STS1	0	0	0	0	0	0	0	0000h
R6273 (1881h)	IRQ1_Raw_Status_2	0	0	CLK_ ASYNC_ ERR_ STS1	CLK_SYS_ ERR_ STS1	0	0	FLL2 LOCK_ STS1	FLL1_ LOCK_ STS1	0	FLL2_ REF_ LOST_ STS1	FLL1_ REF_ LOST_ STS1	0	0	0	0	0	0000h
R6278 (1886h)	IRQ1_Raw_Status_7	0	0	0	0	0	0	0	JD3_STS1	0	0	0	MICD_ CLAMP_ STS1	0	JD2_STS1	0	JD1_STS1	0000h
R6280 (1888h)	IRQ1_Raw_Status_9	0	0	0	0	0	0	ASRC1_ IN2 LOCK_ STS1	ASRC1_ IN1 LOCK_ STS1	0	0	0	0	0	INPUTS_ SIG_DET_ STS1	DRC2_ SIG_DET_ STS1	DRC1 SIG_DET_ STS1	0000h
R6283 (188Bh)	IRQ1_Raw_Status_12	0	0	0	0	0	0	HP4R_ SC_STS1	HP4L_SC_ STS1	0	0	HP3R_ SC_STS1	HP3L_SC_ STS1	HP2R_ SC_STS1	HP2L_SC_ STS1	HP1R_ SC_STS1	HP1L_SC_ STS1	0000h
R6284 (188Ch)	IRQ1_Raw_Status_13	0	0	0	0	0	0	0	0	0	0	HP3R ENABLE_ DONE_ STS1	HP3L ENABLE_ DONE_ STS1	HP2R ENABLE_ DONE_ STS1	HP2L ENABLE_ DONE_ STS1	HP1R ENABLE_ DONE_ STS1	HP1L ENABLE_ DONE_ STS1	0000h
R6285 (188Dh)	IRQ1_Raw_Status_14	0	0	0	0	0	0	0	0	0	0	HP3R_ DISABLE_ DONE_ STS1	DONE_ STS1	HP2R_ DISABLE_ DONE_ STS1	HP2L_ DISABLE_ DONE_ STS1	DONE_ STS1	HP1L_ DISABLE_ DONE_ STS1	0000h
R6288 (1890h)	IRQ1_Raw_Status_17	GP16_ STS1	GP15_ STS1	GP14_ STS1	GP13_ STS1	GP12_ STS1	GP11_ STS1	GP10_ STS1	GP9_STS1	GP8_STS1	GP7_STS1	GP6_STS1	GP5_STS1	GP4_STS1	IGP3_STS1	GP2_STS1	GP1_STS1	0000h
R6292 (1894h)	IRQ1_Raw_Status_21	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	TIMER1_ STS1	0000h
R6293 (1895h)	IRQ1_Raw_Status_22	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ NOT EMPTY_ STS1	0000h
(1896h)	IRQ1_Raw_Status_23	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ FULL_ STS1	0000h
R6295 (1897h)	IRQ1_Raw_Status_24	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ WMARK_ STS1	0000h
R6296 (1898h)	IRQ1_Raw_Status_25	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ DMA_ STS1	0000h
R6301 (189Dh)	IRQ1_Raw_Status_30	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUSY_ STS1	0000h
(18A0h)	IRQ1_Raw_Status_33	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUS_ ERR_ STS1	0000h
R6307 (18A3h)	IRQ1_Raw_Status_36	0	0	0	0	0	0	0	0	0	0	0	0	TIMER_ ALM1_ CH4_STS1	TIMER_ ALM1_ ICH3_STS1	TIMER_ ALM1_ CH2_STS1	TIMER_ ALM1_ CH1_STS1	0000h



Table 6-1.	Register Map	Definition-	-16-bit region	(Cont.)
------------	--------------	-------------	----------------	---------

Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R6311	IRQ1_Raw_Status_40	0	0	0	0	0	0	FLL2	FLL1	0	0	0	0	0	0	0	0	0000h
(18A7h)					OTDUE			SYNC ACTIVE STS1	SYNC ACTIVE STS1	DOOT								00000
R6400 (1900h)	IRQ2_Status_1	0	0	0	CTRLIF_ ERR_ EINT2	0	0	SYSCLK_ FAIL_ EINT2	0	BOOT_ DONE_ EINT2	0	0	0	0	0	0	0	0000h
R6401 (1901h)	IRQ2_Status_2	0	0	CLK ASYNC_ ERR EINT2	CLK_SYS_ ERR EINT2	0	0	FLL2 LOCK EINT2	FLL1_ LOCK_ EINT2	0	FLL2_ REF_ LOST_ EINT2	FLL1_ REF_ LOST_ EINT2	0	0	0	0	0	0000h
R6405 (1905h)	IRQ2_Status_6	0	0	0	0	0	0	MICDET2_ EINT2	MICDET1_ EINT2	0	0	0	0	0	0	0	HPDET_ EINT2	0000h
R6406 (1906h)	IRQ2_Status_7	0	0	0	0	0	0	JD3_ FALL_ EINT2	JD3_ RISE_ EINT2	0	0	MICD_ CLAMP_ FALL_ EINT2	MICD_ CLAMP_ RISE_ EINT2	JD2_ FALL_ EINT2	JD2 RISE EINT2	JD1_ FALL_ EINT2	JD1_ RISE_ EINT2	0000h
R6408 (1908h)	IRQ2_Status_9	0	0	0	0	0	0	ASRC1_ IN2 LOCK_ EINT2	ASRC1_ IN1 LOCK_ EINT2	0	0	0	0	0	INPUTS SIG_DET_ EINT2	DRC2_ SIG_DET_ EINT2	DRC1_ SIG_DET_ EINT2	0000h
R6410 (190Ah)	IRQ2_Status_11	DSP IRQ16 EINT2	DSP IRQ15 EINT2	DSP IRQ14 EINT2	DSP IRQ13_ EINT2	DSP IRQ12 EINT2	DSP IRQ11_ EINT2	DSP IRQ10 EINT2	DSP_ IRQ9_ EINT2	DSP_ IRQ8_ EINT2	DSP_ IRQ7_ EINT2	DSP_ IRQ6_ EINT2	DSP_ IRQ5_ EINT2	DSP_ IRQ4_ EINT2	DSP_ IRQ3_ EINT2	DSP_ IRQ2_ EINT2	DSP_ IRQ1_ EINT2	0000h
R6411 (190Bh)	IRQ2_Status_12	0	0	0	0	0	0	HP4R SC_EINT2	HP4L_SC_ EINT2	0	0	HP3R_ SC_EINT2	HP3L_SC_ EINT2	HP2R SC_EINT2	HP2L_SC_ EINT2	HP1R SC_EINT2	HP1L_SC_ EINT2	0000h
R6412 (190Ch)	IRQ2_Status_13	0	0	0	0	0	0	0	0	0	0	HP3R ENABLE_ DONE_ EINT2	HP3L ENABLE_ DONE_ EINT2	HP2R ENABLE_ DONE_ EINT2	HP2L ENABLE_ DONE_ EINT2	HP1R ENABLE_ DONE_ EINT2	HP1L ENABLE_ DONE_ EINT2	0000h
R6413 (190Dh)	IRQ2_Status_14	0	0	0	0	0	0	0	0	0	0	HP3R_ DISABLE_ DONE_ EINT2	HP3L_ DISABLE_ DONE_ EINT2	HP2R_ DISABLE_ DONE_ EINT2	HP2L_ DISABLE_ DONE_ EINT2	HP1R_ DISABLE_ DONE_ EINT2	HP1L_ DISABLE_ DONE_ EINT2	0000h
R6414 (190Eh)	IRQ2_Status_15	0	0	0	DFC SATURAT E EINT2	0	0	0	0	0	0	0	0	0	0	0	0	0000h
R6416 (1910h)	IRQ2_Status_17	GP16 EINT2	GP15 EINT2	GP14 EINT2	GP13 EINT2	GP12 EINT2	GP11 EINT2	GP10 EINT2	GP9 EINT2	GP8 EINT2	GP7 EINT2	GP6 EINT2	GP5 EINT2	GP4 EINT2	GP3 EINT2	GP2 EINT2	GP1 EINT2	0000h
R6420 (1914h)	IRQ2_Status_21	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	TIMER1_ EINT2	0000h
R6421 (1915h)	IRQ2_Status_22	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ NOT EMPTY_ EINT2	0000h
R6422 (1916h)	IRQ2_Status_23	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ FULL EINT2	0000h
R6423 (1917h)	IRQ2_Status_24	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ WMARK_ EINT2	0000h
R6424 (1918h)	IRQ2_Status_25	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ DMA_ EINT2	0000h
R6426 (191Ah)	IRQ2_Status_27	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ START1_ EINT2	0000h
R6427 (191Bh)	IRQ2_Status_28	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ START2_ EINT2	0000h
R6429 (191Dh)	IRQ2_Status_30	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUSY_ EINT2	0000h
R6432 (1920h)	IRQ2_Status_33	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUS_ ERR_ EINT2	0000h
R6435 (1923h)	IRQ2_Status_36	0	0	0	0	0	0	0	0	0	0	0	0	TIMER_ ALM1_ CH4_ EINT2	TIMER_ ALM1_ CH3_ EINT2	TIMER_ ALM1_ CH2_ EINT2	TIMER_ ALM1_ CH1_ EINT2	0000h
R6439 (1927h)	IRQ2_Status_40	0	0	0	0	0	0	FLL2 SYNC ACTIVE_ EINT2	FLL1_ SYNC_ ACTIVE_ EINT2	0	0	0	0	0	0	0	0	0000h
R6464 (1940h)	IRQ2_Mask_1	0	0	0	IM CTRLĪF_ ERR_ EINT2	0	0	IM SYSCLK_ FAIL_ EINT2	0	IM_BOOT_ DONE_ EINT2	0	0	0	0	0	0	0	1280h
R6465 (1941h)	IRQ2_Mask_2	0	1	IM_CLK_ ASYNC_ ERR_ EINT2	IM_CLK_ SYS_ ERR_ EINT2	0	0	IM_FLL2_ LOCK_ EINT2	IM_FLL1_ LOCK_ EINT2	0	IM_FLL2_ REF_ LOST_ EINT2	IM_FLL1_ REF_ LOST_ EINT2	0	0	0	0	0	77E0h
R6469 (1945h)	IRQ2_Mask_6	0	0	0	0	0	0	IM_ MICDET2_ EINT2	IM_ MICDET1_ EINT2	0	0	0	0	0	0	0	IM HPDET_ EINT2	0301h
R6470 (1946h)	IRQ2_Mask_7	0	0	0	0	0	0	IM_JD3_ FALL EINT2	IM_JD3_ RISE_ EINT2	0	0	IM_MICD_ CLAMP_ FALL EINT2	IM_MICD_ CLAMP_ RISE_ EINT2	IM_JD2_ FALL EINT2	IM_JD2_ RISE_ EINT2	IM_JD1_ FALL EINT2	IM_JD1_ RISE EINT2	033Fh



Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R6472 (1948h)	IRQ2_Mask_9	0	0	0	0	0	0	IM ASRC1	IM ASRC1	0	0	0	1	1	IM_ INPUTS		IM_DRC1_ SIG_DET_	031Fh
(19401)								IN2 LOCK EINT2	IN1 LOCK EINT2						SIG_DET_ EINT2	EINT2	EINT2	
R6474 (194Ah)	IRQ2_Mask_11	IM_DSP_ IRQ16_ EINT2	IM_DSP_ IRQ15_ EINT2	IM_DSP_ IRQ14_ EINT2	IM_DSP_ IRQ13_ EINT2	IM_DSP_ IRQ12_ EINT2	IM_DSP_ IRQ11_ EINT2	IM_DSP_ IRQ10_ EINT2	IM_DSP_ IRQ9_ EINT2	IM_DSP_ IRQ8_ EINT2	IM_DSP_ IRQ7_ EINT2	IM_DSP_ IRQ6_ EINT2	IM_DSP_ IRQ5_ EINT2	IM_DSP_ IRQ4_ EINT2	IM_DSP_ IRQ3_ EINT2	IM_DSP_ IRQ2_ EINT2	IM_DSP_ IRQ1_ EINT2	FFFFh
R6475 (194Bh)	IRQ2_Mask_12	0	0	0	0	0	0		IM_HP4L SC_EINT2	0	0	IM_HP3R_ SC_EINT2	IM_HP3L SC_EINT2	IM_HP2R_ SC_EINT2	IM_HP2L SC_EINT2	IM_HP1R_ SC_EINT2	IM_HP1L SC_EINT2	033Fh
R6476 (194Ch)	IRQ2_Mask_13	0	0	0	0	0	0	0	0	0	0	IM_HP3R_ ENABLE_ DONE_	IM_HP3L_ ENABLE_ DONE_	IM_HP2R_ ENABLE_ DONE_ EINT2	IM_HP2L_ ENABLE_ DONE_	IM_HP1R_ ENABLE_ DONE_ EINT2	IM_HP1L_ ENABLE_ DONE_	003Fh
R6477 (194Dh)	IRQ2_Mask_14	0	0	0	0	0	0	0	0	0	0	EINT2 IM_HP3R_ DISABLE_ DONE_	EINT2 IM_HP3L_ DISABLE_ DONE_	IM_HP2R_ DISABLE_ DONE	EINT2 IM_HP2L_ DISABLE_ DONE_	IM_HP1R_ DISABLE_ DONE_	EINT2 IM_HP1L_ DISABLE_ DONE_	003Fh
R6478 (194Eh)	IRQ2_Mask_15	0	0	0	IM_DFC_ SATURAT E_EINT2	0	0	0	0	0	0	EINT2 0	EINT2 0	EINT2 0	EINT2 0	EINT2 0	EINT2 0	1000h
R6480 (1950h)	IRQ2_Mask_17	IM_GP16_ EINT2	IM_GP15_ EINT2	IM_GP14_ EINT2		IM_GP12_ EINT2	IM_GP11_ EINT2	IM_GP10_ EINT2	IM_GP9_ EINT2	IM_GP8_ EINT2	IM_GP7_ EINT2	IM_GP6_ EINT2	IM_GP5_ EINT2	IM_GP4_ EINT2	IM_GP3_ EINT2	IM_GP2_ EINT2	IM_GP1_ EINT2	FFFFh
R6484 (1954h)	IRQ2_Mask_21	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_ TIMER1_ EINT2	0001h
R6485 (1955h)	IRQ2_Mask_22	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_ EVENT1_ NOT_ EMPTY_ EINT2	0001h
R6486 (1956h)	IRQ2_Mask_23	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_ EVENT1_ FULL EINT2	0001h
R6487 (1957h)	IRQ2_Mask_24	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_ EVENT1_ WMARK_ EINT2	0001h
R6488 (1958h)	IRQ2_Mask_25	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ DMA_ EINT2	0001h
R6490 (195Ah)	IRQ2_Mask_27	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ START1_ EINT2	0001h
R6491 (195Bh)	IRQ2_Mask_28	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ START2_ EINT2	0001h
R6493 (195Dh)	IRQ2_Mask_30	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ BUSY_ EINT2	0001h
R6496 (1960h)	IRQ2_Mask_33	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	IM_DSP1_ BUS_ ERR_ EINT2	0001h
R6499 (1963h)	IRQ2_Mask_36	0	0	0	0	0	0	0	0	0	0	0	0	IM_ TIMER_ ALM1_ CH4_ EINT2	IM_ TIMER_ ALM1_ CH3_ EINT2	IM_ TIMER_ ALM1_ CH2_ EINT2	IM_ TIMER_ ALM1_ CH1_ EINT2	000Fh
R6503 (1967h)	IRQ2_Mask_40	0	0	0	0	0	0	IM_FLL2_ SYNC ACTIVE_ EINT2	IM_FLL1_ SYNC ACTIVE_ EINT2	0	0	0	0	0	0	0	0	0300h
R6528 (1980h)	IRQ2_Raw_Status_1	0	0	0	CTRLIF_ ERR STS2	0	0	0	0	BOOT_ DONE_ STS2	0	0	0	0	0	0	0	0000h
R6529 (1981h)	IRQ2_Raw_Status_2	0	0	CLK ASYNC_ ERR_ STS2	CLK_SYS_ ERR STS2	0	0	FLL2_ LOCK_ STS2	FLL1 LOCK_ STS2	0	FLL2_ REF_ LOST_ STS2	FLL1_ REF_ LOST_ STS2	0	0	0	0	0	0000h
R6534 (1986h)	IRQ2_Raw_Status_7	0	0	0	0	0	0	0	JD3_STS2	0	0	0	MICD_ CLAMP_ STS2	0	JD2_STS2	0	JD1_STS2	0000h
R6536 (1988h)	IRQ2_Raw_Status_9	0	0	0	0	0	0	ASRC1_ IN2 LOCK_ STS2	ASRC1_ IN1 LOCK_ STS2	0	0	0	0	0	INPUTS SIG_DET_ STS2	DRC2 SIG_DET_ STS2	DRC1_ SIG_DET_ STS2	0000h
R6539 (198Bh)	IRQ2_Raw_Status_12	0	0	0	0	0	0	HP4R SC_STS2	HP4L_SC_ STS2	0	0	HP3R_ SC_STS2	HP3L_SC_ STS2	HP2R_ SC_STS2	HP2L_SC_ STS2	HP1R_ SC_STS2	HP1L_SC_ STS2	0000h
R6540 (198Ch)	IRQ2_Raw_Status_13	0	0	0	0	0	0	0	0	0	0	HP3R ENABLE_ DONE_ STS2	HP3L ENABLE_ DONE_ STS2	HP2R ENABLE_ DONE_ STS2	HP2L ENABLE_ DONE_ STS2	HP1R ENABLE_ DONE_ STS2	HP1L ENABLE_ DONE_ STS2	0000h
R6541 (198Dh)	IRQ2_Raw_Status_14	0	0	0	0	0	0	0	0	0	0	HP3R DISABLE_ DONE_ STS2	HP3L_ DISABLE_ DONE_ STS2	HP2R DISABLE_ DONE_ STS2	HP2L_ DISABLE_ DONE_ STS2	HP1R DISABLE_ DONE_ STS2	HP1L_ DISABLE_ DONE_ STS2	0000h
R6544 (1990h)	IRQ2_Raw_Status_17	GP16_ STS2	GP15_ STS2	GP14_ STS2	GP13_ STS2	GP12_ STS2	GP11_ STS2	GP10_ STS2	GP9_STS2	GP8_STS2	GP7_STS	GP6_STS2					GP1_STS2	0000h
R6548 (1994h)	IRQ2_Raw_Status_21	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	TIMER1_ STS2	0000h



Register	Name	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	Default
R6549 (1995h)	IRQ2_Raw_Status_22	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ NOT EMPTY_ STS2	0000h
R6550 (1996h)	IRQ2_Raw_Status_23	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ FULL_ STS2	0000h
R6551 (1997h)	IRQ2_Raw_Status_24	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVENT1_ WMARK_ STS2	0000h
R6552 (1998h)	IRQ2_Raw_Status_25	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ DMA_ STS2	0000h
R6557 (199Dh)	IRQ2_Raw_Status_30	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUSY_ STS2	0000h
R6560 (19A0h)	IRQ2_Raw_Status_33	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ BUS_ ERR_ STS2	0000h
R6563 (19A3h)	IRQ2_Raw_Status_36	0	0	0	0	0	0	0	0	0	0	0	0	TIMER_ ALM1_ CH4_STS2	TIMER_ ALM1_ CH3_STS2	TIMER_ ALM1_ CH2_STS2	TIMER_ ALM1_ CH1_STS2	0000h
R6567 (19A7h)	IRQ2_Raw_Status_40	0	0	0	0	0	0	FLL2 SYNC ACTIVE_ STS2	FLL1_ SYNC_ ACTIVE_ STS2	0	0	0	0	0	0	0	0	0000h
R6662 (1A06h)	Interrupt_Debounce_7	0	0	0	0	0	0	0	JD3_DB	0	0	0	MICD_ CLAMP_ DB	0	JD2_DB	0	JD1_DB	0000h
R6784 (1A80h)	IRQ1_CTRL	0	1	0	0	IM_IRQ1	IRQ_POL	IRQ_OP_ CFG	0	0	0	0	0	0	0	0	0	4400h
R6786 (1A82h)	IRQ2_CTRL	0	0	0	0	IM_IRQ2	0	0	0	0	0	0	0	0	0	0	0	0000h
R6816 (1AA0h)	Interrupt_Raw_Status_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	_	IRQ1_STS	0000h
R6848 (1AC0h)	GPIO_Debounce_Config	0	0	0	0	0	0	0	0	0	0	0	0		GP_DBT	TME [3:0]		0000h
R6864 (1AD0h)	AOD_Pad_Ctrl	0	1	0	0	0	0	0	0	0	0	0	0	0	0	RESET_ PU	RESET_ PD	4002h

The 32-bit register space is described in Table 6-2.

Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12288	WSEQ_Sequence_1	WSEQ_D/	ATA_WIDT	H0 [2:0]						WSE	Q_ADDR0	[12:0]						0000F000h
(3000h)		۷	NSEQ_DE	LAY0 [3:0]		W	SEQ_DATA	_START0[3:0]				WSEQ_	DATA0 [7:0]				
R12290	WSEQ_Sequence_2	WSEQ_D/	_							WSE	Q_ADDR1	[12:0]						0000F000h
(3002h)			_	LAY1 [3:0]		W	SEQ_DATA	_START1[3:0]				WSEQ_	DATA1 [7:0]				
R12292	WSEQ_Sequence_3	WSEQ_D/	_							WSE	Q_ADDR2	[12:0]						0000F000h
(3004h)				LAY2 [3:0]		W	SEQ_DATA	_START2[3:0]				WSEQ_	DATA2 [7:0]				
R12294	WSEQ_Sequence_4	WSEQ_D/	_							WSE	Q_ADDR3	[12:0]						0000F000h
(3006h)				LAY3 [3:0]		W	SEQ_DATA	_START3 [3:0]				WSEQ_	DATA3 [7:0]				
R12296	WSEQ_Sequence_5	WSEQ_D/	ATA_WIDT	H4 [2:0]						WSE	Q_ADDR4	[12:0]						82253719h
(3008h)				LAY4 [3:0]		W	SEQ_DATA	_START4 [3:0]				WSEQ_	DATA4 [7:0]				
R12298	WSEQ_Sequence_6	WSEQ_D/	ATA_WIDT	H5 [2:0]						WSE	Q_ADDR5	[12:0]						C2300001h
(300Ah)		۷	NSEQ_DE	LAY5 [3:0]		W	SEQ_DATA	_START5 [3:0]				WSEQ_	DATA5 [7:0]				
R12300	WSEQ_Sequence_7	WSEQ_D/	ATA_WIDT	H6 [2:0]						WSE	Q_ADDR6	[12:0]						02251301h
(300Ch)				LAY6 [3:0]		W	SEQ_DATA	_START6[3:0]				WSEQ_	DATA6 [7:0]				
R12302	WSEQ_Sequence_8	WSEQ_D/	_							WSE	Q_ADDR7	[12:0]						8225191Fh
(300Eh)		۷	NSEQ_DE	LAY7 [3:0]		W	SEQ_DATA	_START7 [3:0]				WSEQ_	DATA7 [7:0]				
R12304	WSEQ_Sequence_9	WSEQ_D/	ATA_WIDT	H8 [2:0]		_				WSE	Q_ADDR8	[12:0]						82310B00h
(3010h)		۷	NSEQ_DE	LAY8 [3:0]		W	SEQ_DATA	_START8 [3:0]				WSEQ_	DATA8 [7:0]				
R12306	WSEQ_Sequence_10	WSEQ_D/	ATA_WIDT	H9 [2:0]						WSE	Q_ADDR9	[12:0]						E231023Bh
(3012h)		۷	NSEQ_DE	LAY9 [3:0]		W	SEQ_DATA	_START9 [3:0]				WSEQ_	DATA9 [7:0]				
R12308	WSEQ_Sequence_11	WSEQ_DA	ATA_WIDT	H10 [2:0]		_				WSE	ADDR10_	[12:0]						02313B01h
(3014h)		N	VSEQ_DEI	_AY10 [3:0]		WS	SEQ_DATA	_START10	[3:0]				WSEQ_I	DATA10 [7:0]				
R12310	WSEQ_Sequence_12	WSEQ_DA	ATA_WIDT	H11 [2:0]						WSE	ຊ_ADDR11	[12:0]						62300000h
(3016h)		W	VSEQ_DEI	LAY11 [3:0]		WS	SEQ_DATA	START11	[3:0]				WSEQ_I	DATA11 [7:0]				
R12312	WSEQ_Sequence_13	WSEQ_DA	ATA_WIDT	H12 [2:0]						WSE	ADDR12_ADDR12	[12:0]						E2314288h
(3018h)		N	VSEQ_DEI	_AY12 [3:0]		WS	SEQ_DATA	START12	[3:0]				WSEQ_I	DATA12 [7:0]				
R12314	WSEQ_Sequence_14	WSEQ_DA	ATA_WIDT	H13 [2:0]						WSE	ADDR13_	[12:0]						02310B00h
(301Ah)		N	VSEQ_DEI	_AY13 [3:0]		WS	SEQ_DATA	START13	[3:0]				WSEQ_I	DATA13 [7:0]				
R12316	WSEQ_Sequence_15	WSEQ_DA	ATA_WIDT	H14 [2:0]						WSE	ADDR14_	[12:0]						02310B00h
(301Ch)		N	VSEQ_DEI	_AY14 [3:0]		WS	SEQ_DATA	_START14	[3:0]				WSEQ_I	DATA14 [7:0]				1
R12318	WSEQ_Sequence_16	WSEQ_DA	TA_WIDT	H15 [2:0]						WSE	ADDR15_	[12:0]						02250E01h
(301Eh)		W	VSEQ_DEI	AY15 [3:0]		WS	SEQ_DATA	START15	[3:0]				WSEQ_I	DATA15 [7:0]				



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12320 (3020h)	WSEQ_Sequence_17		ATA_WIDT	H16 [2:0] LAY16 [3:0]		WS	SEQ_DATA	START16	\$ [3:0]	WSE	2_ADDR16	[12:0]	WSEO D	ATA16 [7:0	1			42310C02h
R12322 (3022h)	WSEQ_Sequence_18	WSEQ_D	ATA_WIDT				SEQ_DATA	_		WSE	2_ADDR17	[12:0]		ATA17 [7:0	-			E2310227h
R12324 (3024h)	WSEQ_Sequence_19	WSEQ_D	ATA_WIDT				SEQ_DATA			WSE	Q_ADDR18	[12:0]	_	ATA18 [7:0				02313B01h
R12326 (3026h)	WSEQ_Sequence_20	_	ATA_WIDT					_		WSE	ADDR19	[12:0]	-	47440 17.0				E2314266h
R12328 (3028h)	WSEQ_Sequence_21	WSEQ_D	ATA_WIDT	LAY19 [3:0] H20 [2:0] LAY20 [3:0]			SEQ_DATA	_		WSE	Q_ADDR20	[12:0]		ATA19 [7:0				E2315294h
R12330 (302Ah)	WSEQ_Sequence_22	WSEQ_D	ATA_WIDT				SEQ_DATA	_		WSE	Q_ADDR21	[12:0]	_	ATA20 [7:0	-			02310B00h
R12332 (302Ch)	WSEQ_Sequence_23	WSEQ_D	ATA_WIDT				SEQ DATA	_		WSE	Q_ADDR22	[12:0]		ATA22 [7:0	-			02310B00h
R12334 (302Eh)	WSEQ_Sequence_24		ATA_WIDT WSEQ_DE	H23 [2:0] LAY23 [3:0]		WS	SEQ_DATA	_START23	8 [3:0]	WSE	Q_ADDR23	[12:0]	WSEQ_D	ATA23 [7:0]			E2251734h
R12336 (3030h)	WSEQ_Sequence_25	1		LAY24 [3:0]		WS	SEQ_DATA	_START24	l [3:0]	WSE	2_ADDR24	[12:0]	WSEQ_D	ATA24 [7:0]			0225F501h
R12338 (3032h)	WSEQ_Sequence_26			LAY25 [3:0]		WS	SEQ_DATA	_START25	5 [3:0]		Q_ADDR25		WSEQ_D	ATA25 [7:0]			0000F000h
R12340 (3034h)	WSEQ_Sequence_27			LAY26 [3:0]		WS	SEQ_DATA	_START26	6 [3:0]		2_ADDR26		WSEQ_D	ATA26 [7:0]			0000F000h
R12342 (3036h)	WSEQ_Sequence_28		-	LAY27 [3:0]		WS	SEQ_DATA	_START27	7 [3:0]		Q_ADDR27		WSEQ_D	ATA27 [7:0]			0000F000h
R12344 (3038h)	WSEQ_Sequence_29			LAY28 [3:0]		WS	SEQ_DATA	_START28	8 [3:0]		Q_ADDR28		WSEQ_D	ATA28 [7:0]			0000F000h
R12346 (303Ah)	WSEQ_Sequence_30	1	_	LAY29 [3:0]		WS	SEQ_DATA	_START29	9 [3:0]		Q_ADDR29		WSEQ_D	ATA29 [7:0]			0000F000h
R12348 (303Ch)	WSEQ_Sequence_31	1	ATA_WIDT WSEQ_DE ATA WIDT	LAY30 [3:0]		WS	SEQ_DATA	_START30) [3:0]		Q_ADDR30		WSEQ_D	ATA30 [7:0]			0000F000h
R12350 (303Eh)	WSEQ_Sequence_32		WSEQ_DE	LAY31 [3:0]		WS	SEQ_DATA	_START31	[3:0]		-		WSEQ_D	ATA31 [7:0]			02253A01h
R12352 (3040h) R12354	WSEQ_Sequence_33		ATA_WIDT WSEQ_DE ATA_WIDT	LAY32 [3:0]		WS	SEQ_DATA	_START32	2 [3:0]		Q_ADDR32		WSEQ_D	ATA32 [7:0]			C2251300h
(3042h) R12356	WSEQ_Sequence_34		_	LAY33 [3:0]		WS	SEQ_DATA	_START33	8 [3:0]		2_ADDR33		WSEQ_D	ATA33 [7:0]			02250B00h
(3044h) R12358	WSEQ_Sequence_35 WSEQ_Sequence_36		_	LAY34 [3:0]		WS	SEQ_DATA	_START34	l [3:0]		2 ADDR35		WSEQ_D	ATA34 [7:0]			0225FF011
(3046h) R12360	WSEQ Sequence 37		_	LAY35 [3:0]		WS	SEQ_DATA	_START35	5 [3:0]		Q ADDR36		WSEQ_D	ATA35 [7:0]			0000F000h
(3048h) R12362	WSEQ_Sequence_38			LAY36 [3:0]		WS	SEQ_DATA	_START36	6 [3:0]		Q ADDR37		WSEQ_D	ATA36 [7:0]			0000F000ł
(304Ah) R12364	WSEQ_Sequence 39	1		LAY37 [3:0]		WS	SEQ_DATA	_START37	7 [3:0]		Q ADDR38		WSEQ_D	ATA37 [7:0]			0000F000h
(304Ch) R12366	WSEQ_Sequence_39	1		LAY38 [3:0]		WS	SEQ_DATA	_START38	8 [3:0]		Q ADDR39		WSEQ_D	ATA38 [7:0]			0000F000h
(304Eh)	WSEQ_Sequence_40	1		LAY39 [3:0]		WS	SEQ_DATA	_START39	9 [3:0]		Q ADDR40		WSEQ_D	ATA39 [7:0]			0000F000h
(3050h) R12370	WSEQ_Sequence 42		_	LAY40 [3:0]		WS	SEQ_DATA	_START40	0 [3:0]		Q ADDR41		WSEQ_D	ATA40 [7:0]			0000F000h
(3052h)	WSEQ_Sequence_43	1	_	LAY41 [3:0]		WS	SEQ_DATA	_START41	[3:0]		2_ADDR42		WSEQ_D	ATA41 [7:0]			0000F000h
(3054h)	WSEQ_Sequence_44	1		LAY42 [3:0]		WS	SEQ_DATA	_START42	2 [3:0]		Q ADDR43		WSEQ_D	ATA42 [7:0]			0000F000h
(3056h) R12376	WSEQ Sequence 45	1		LAY43 [3:0]		WS	SEQ_DATA	_START43	8 [3:0]		Q ADDR44		WSEQ_D	ATA43 [7:0]			82263719h
(3058h)	WSEQ_Sequence_46	1	_	LAY44 [3:0]		WS	SEQ_DATA	_START44	l [3:0]		Q ADDR45		WSEQ_D	ATA44 [7:0]			C2300001h
(305Ah) R12380	WSEQ_Sequence_47	1	_	LAY45 [3:0]		WS	SEQ_DATA	_START45	5 [3:0]		Q ADDR46		WSEQ_D	ATA45 [7:0]			02261301h
(305Ch) R12382	WSEQ_Sequence_48	1		LAY46 [3:0]		WS	SEQ_DATA	_START46	6 [3:0]		2_ADDR47		WSEQ_D	ATA46 [7:0]			8226191Fh
(305Eh) R12384	WSEQ_Sequence_49	1		LAY47 [3:0]		WS	SEQ_DATA	_START47	7 [3:0]		2_ADDR48		WSEQ_D	ATA47 [7:0]			82310B02h
(3060h) R12386	WSEQ Sequence 50	1		LAY48 [3:0]		WS	SEQ_DATA	_START48	8 [3:0]		Q_ADDR49		WSEQ_D	ATA48 [7:0]			E231023B
(3062h) R12388	WSEQ_Sequence_51		_	LAY49 [3:0]		WS	SEQ_DATA	_START49	9 [3:0]		2_ADDR50		WSEQ_D	ATA49 [7:0]			02313B01h
(3064h) R12390	WSEQ_Sequence_52	1		LAY50 [3:0]		WS	SEQ_DATA	_START50) [3:0]		2_ADDR51		WSEQ_D	ATA50 [7:0]			62300000h
(3066h)				LAY51 [3:0]		WS	SEQ_DATA	_START51	[3:0]				WSEQ_D	ATA51 [7:0]			



Register	Name			29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12392 (3068h)	WSEQ_Sequence_53	WSEQ_DATA_	WIDTH52			WS	EQ DATA	START52	3:01	WSE	2_ADDR52	[12:0]	WSEQ D	ATA52 [7:0]	1			E2314288h
R12394 (306Ah)	WSEQ_Sequence_54	WSEQ_DATA_		3 [2:0]				START53		WSE	Q_ADDR53	[12:0]	-	ATA53 [7:0]				02310B00h
R12396 (306Ch)	WSEQ_Sequence_55	WSEQ_DATA_		4 [2:0]				START54		WSE	Q_ADDR54	[12:0]		ATA54 [7:0]				02310B00h
R12398 (306Eh)	WSEQ_Sequence_56	WSEQ_DATA_	_WIDTH55	5 [2:0]			_			WSE	Q_ADDR55	[12:0]						02260E01h
R12400 (3070h)	WSEQ_Sequence_57	WSEQ_DATA_	Q_DELAY _WIDTH56 Q_DELAY	6 [2:0]				START55 [WSE	Q_ADDR56	[12:0]		ATA55 [7:0] ATA56 [7:0]				42310C03h
R12402 (3072h)	WSEQ_Sequence_58	WSEQ_DATA_	_	7 [2:0]			_	START57		WSE	Q_ADDR57	[12:0]	_	ATA57 [7:0]				E2310227h
R12404 (3074h)	WSEQ_Sequence_59	WSEQ_DATA_	_WIDTH58	3 [2:0]	1		_			WSE	Q_ADDR58	[12:0]	_					02313B01h
R12406	WSEQ_Sequence_60	WSEQ_DATA_	-	9 [2:0]				START58 [WSE	Q_ADDR59	[12:0]		ATA58 [7:0]				E2314266h
(3076h) R12408	WSEQ_Sequence_61	WSEQ_DATA_	Q_DELAY _WIDTH60			WS	EQ_DATA	_START59 [3:0]	WSE	Q_ADDR60	[12:0]	WSEQ_D	ATA59 [7:0]				E2315294h
(3078h) R12410	WSEQ Sequence 62	WSEQ_DATA	Q_DELAY WIDTH61			WS	EQ_DATA	_START60 [3:0]	WSE	2_ADDR61	[12:0]	WSEQ_D	ATA60 [7:0]				02310B00h
(307Ah) R12412	WSEQ Sequence 63	WSEQ_DATA	Q_DELAY WIDTH62			WS	EQ_DATA	_START61 [3:0]	WSE	Q ADDR62	[12:0]	WSEQ_D	ATA61 [7:0]				02310B00h
(307Ch) R12414		WSE	Q_DELAY	62 [3:0]		WS	EQ_DATA	_START62 [3:0]		-		WSEQ_D	ATA62 [7:0]				
(307Eh)	WSEQ_Sequence_64		- Q_DELAY	63 [3:0]		WS	EQ_DATA	_START63 [3:0]		Q_ADDR63		WSEQ_D	ATA63 [7:0]				E2261734h
R12416 (3080h)	WSEQ_Sequence_65		- Q_DELAY	64 [3:0]		WS	EQ_DATA	_START64 [3:0]	WSE	2_ADDR64	[12:0]	WSEQ_D	ATA64 [7:0]				0226F501h
R12418 (3082h)	WSEQ_Sequence_66	WSEQ_DATA_	_WIDTH65 Q_DELAY			WS	EQ_DATA	START65 [3:0]	WSE	Q_ADDR65	[12:0]	WSEQ_D	ATA65 [7:0]				0000F000h
R12420 (3084h)	WSEQ_Sequence_67	WSEQ_DATA_	WIDTH66			WS	FO DATA	START66 [3.01	WSE	Q_ADDR66	[12:0]	WSEQ D	ATA66 [7:0]				0000F000h
R12422 (3086h)	WSEQ_Sequence_68	WSEQ_DATA_	-	7 [2:0]			_	START67 [WSE	Q_ADDR67	[12:0]		ATA67 [7:0]				0000F000h
R12424 (3088h)	WSEQ_Sequence_69	WSEQ_DATA_	WIDTH68	8 [2:0]						WSE	Q_ADDR68	[12:0]						0000F000h
R12426	WSEQ_Sequence_70	WSEQ_DATA_		9 [2:0]				_START68 [WSE	Q_ADDR69	[12:0]		ATA68 [7:0]				0000F000h
(308Ah) R12428	WSEQ_Sequence_71	WSEQ_DATA_	Q_DELAY WIDTH70			WS	EQ_DATA	_START69 [3:0]	WSE	Q_ADDR70	[12:0]	WSEQ_D	ATA69 [7:0]				0000F000h
(308Ch) R12430	WSEQ_Sequence_72	WSEQ_DATA_	Q_DELAY WIDTH71			WS	EQ_DATA	_START70 [3:0]	WSE	Q_ADDR71	[12:0]	WSEQ_D	ATA70 [7:0]				02263A01h
(308Eh) R12432	WSEQ Sequence 73	WSEQ DATA	Q_DELAY WIDTH72			WS	EQ_DATA	_START71 [3:0]	WSE	Q ADDR72	[12:0]	WSEQ_D	ATA71 [7:0]				C2261300h
(3090h) R12434	WSEQ Sequence 74		- Q_DELAY	72 [3:0]		WS	EQ_DATA	_START72 [3:0]		Q ADDR73		WSEQ_D	ATA72 [7:0]				02260B00h
(3092h)		WSE	- Q_DELAY	73 [3:0]		WS	EQ_DATA	_START73 [3:0]		-		WSEQ_D	ATA73 [7:0]				
R12436 (3094h)	WSEQ_Sequence_75		- Q_DELAY	74 [3:0]		WS	EQ_DATA	_START74 [3:0]		Q_ADDR74		WSEQ_D	ATA74 [7:0]				0226FF01h
R12438 (3096h)	WSEQ_Sequence_76	WSEQ_DATA_ WSE0	_WIDTH75 Q_DELAY			WS	EQ_DATA	_START75 [3:0]	WSE	Q_ADDR75	[12:0]	WSEQ_D	ATA75 [7:0]				0000F000h
R12440 (3098h)	WSEQ_Sequence_77	WSEQ_DATA_ WSE0	_WIDTH76 Q DELAY			WS	EQ DATA	START76	3:0]	WSE	Q_ADDR76	[12:0]	WSEQ D	ATA76 [7:0]				0000F000h
R12442 (309Ah)	WSEQ_Sequence_78	WSEQ_DATA_	-	7 [2:0]			_	START77 [WSE	Q_ADDR77	[12:0]	_	ATA77 [7:0]				0000F000h
R12444 (309Ch)	WSEQ_Sequence_79	WSEQ_DATA_	WIDTH78	8 [2:0]			_			WSE	Q_ADDR78	[12:0]	-					0000F000h
R12446	WSEQ_Sequence_80	WSEQ_DATA_		9 [2:0]			_	_START78 [WSE	Q_ADDR79	[12:0]		ATA78 [7:0]				0000F000h
(309Eh) R12448	WSEQ_Sequence_81	WSEQ_DATA_	Q_DELAY WIDTH80			WS	EQ_DATA	_START79 [3:0]	WSE	Q_ADDR80	[12:0]	_	ATA79 [7:0]				0000F000h
(30A0h) R12450	WSEQ_Sequence_82	WSEQ_DATA_	Q_DELAY WIDTH81			WS	eq_data	_START80 [3:0]	WSE	2_ADDR81	[12:0]	WSEQ_D	ATA80 [7:0]				0000F000h
(30A2h) R12452	WSEQ_Sequence_83		Q_DELAY	'81 [3:0]		WS	eq_data	_START81 [3:0]		Q ADDR82		WSEQ_D	ATA81 [7:0]				0000F000h
(30A4h)		WSE	Q_DELAY	82 [3:0]		WS	eq_data	_START82 [3:0]		-		WSEQ_D	ATA82 [7:0]				
R12454 (30A6h)	WSEQ_Sequence_84		Q_DELAY	83 [3:0]		WS	EQ_DATA	_START83 [3:0]		Q_ADDR83		WSEQ_D	ATA83 [7:0]				0000F000h
R12456 (30A8h)	WSEQ_Sequence_85	WSEQ_DATA_ WSE0	_WIDTH84 Q_DELAY			WS	EQ_DATA	_START84 [3:0]	WSE	Q_ADDR84	[12:0]	WSEQ_D	ATA84 [7:0]				82273719h
R12458 (30AAh)	WSEQ_Sequence_86	WSEQ_DATA_		5 [2:0]				START85 [WSE	Q_ADDR85	[12:0]	WSEQ D	ATA85 [7:0]				C2400001h
R12460 (30ACh)	WSEQ_Sequence_87	WSEQ_DATA_	_WIDTH86	6 [2:0]	I					WSE	Q_ADDR86	[12:0]		ATA86 [7:0]				02271301h
R12462	WSEQ_Sequence_88	WSEQ_DATA_		7 [2:0]			_	START86 [-	WSE	Q_ADDR87	[12:0]						8227191Fh
(30AEh)		WSE	Q_DELAY	87 [3:0]		WS	EQ_DATA	_START87 [3:0]				WSEQ_D	ATA87 [7:0]				<u> </u>



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12464 (30B0h)	WSEQ_Sequence_89		ATA_WIDT WSEQ DE			18/0		START88	[3.0]	WSEC	2_ADDR88	[12:0]	WSEO P	ATA88 [7:0				82410B00h
R12466 (30B2h)	WSEQ_Sequence_90	WSEQ_D	ATA_WIDT				_	START89		WSEC	_ADDR89	[12:0]		ATA89 [7:0	-			E241023Bh
R12468 (30B4h)	WSEQ_Sequence_91	WSEQ_D	ATA_WIDT				_	START90		WSEC	2_ADDR90	[12:0]	_	ATA90 [7:0				02413B01h
R12470 (30B6h)	WSEQ_Sequence_92	-	ATA_WIDT			10/0		OTA DTO1	[2,0]	WSEC	2_ADDR91	[12:0]		ATA04 [7.0	1			62400000h
R12472 (30B8h)	WSEQ_Sequence_93	WSEQ_D	ATA_WIDT	LAY91 [3:0] H92 [2:0] LAY92 [3:0]				_START91		WSEC	_ADDR92	[12:0]		ATA91 [7:0				E2414288h
R12474 (30BAh)	WSEQ_Sequence_94	WSEQ_D	ATA_WIDT				_	_START93		WSEC	2_ADDR93	[12:0]	-	ATA93 [7:0				02410B00h
R12476 (30BCh)	WSEQ_Sequence_95		ATA_WIDT	H94 [2:0] LAY94 [3:0]		WS	EQ_DATA	START94	[3:0]	WSEC	2_ADDR94	[12:0]	WSEQ_D	ATA94 [7:0]			02410B00h
R12478 (30BEh)	WSEQ_Sequence_96	1		LAY95 [3:0]		WS	SEQ_DATA	_START95	[3:0]		ADDR95		WSEQ_D	ATA95 [7:0]			02270E01h
R12480 (30C0h)	WSEQ_Sequence_97	1		LAY96 [3:0]		WS	SEQ_DATA	_START96	[3:0]		ADDR96		WSEQ_D	ATA96 [7:0]			42410C02h
R12482 (30C2h)	WSEQ_Sequence_98		ATA_WIDT WSEQ_DE	H97 [2:0] LAY97 [3:0]		WS	SEQ_DATA	_START97	[3:0]	WSEC	2_ADDR97	[12:0]	WSEQ_D	ATA97 [7:0]			E2410227h
R12484 (30C4h)	WSEQ_Sequence_99			LAY98 [3:0]		WS	SEQ_DATA	_START98	[3:0]		2_ADDR98		WSEQ_D	ATA98 [7:0]			02413B01h
R12486 (30C6h)	WSEQ_Sequence_100		_	LAY99 [3:0]		WS	SEQ_DATA	_START99	[3:0]		2_ADDR99		WSEQ_D	ATA99 [7:0]			E2414266h
R12488 (30C8h)	WSEQ_Sequence_101	V		AY100 [3:0]	WS	EQ_DATA	START100) [3:0]		_ADDR100		WSEQ_D	ATA100 [7:0)]			E2415294h
R12490 (30CAh)	WSEQ_Sequence_102	V	_	AY101 [3:0]	WS	EQ_DATA	START101	[3:0]		_ADDR101		WSEQ_D	ATA101 [7:0)]			02410B00h
R12492 (30CCh)	WSEQ_Sequence_103	V		AY102 [3:0]	WS	EQ_DATA	_START102	2 [3:0]		_ADDR102		WSEQ_D	ATA102 [7:0)]			02410B00h
R12494 (30CEh)	WSEQ_Sequence_104	V	_	AY103 [3:0]	WS	EQ_DATA	START103	8 [3:0]		_ADDR103		WSEQ_D	ATA103 [7:0	0]			E2271734h
R12496 (30D0h)	WSEQ_Sequence_105	V		AY104 [3:0]	WS	EQ_DATA	_START104	[3:0]		_ADDR104		WSEQ_D	ATA104 [7:0)]			0227F501h
R12498 (30D2h)	WSEQ_Sequence_106	V	_	AY105 [3:0]	WS	EQ_DATA	START105	5 [3:0]		_ADDR105		WSEQ_D	ATA105 [7:0)]			0000F000h
R12500 (30D4h)	WSEQ_Sequence_107	V		AY106 [3:0]	WS	EQ_DATA	_START106	6 [3:0]	1	_ADDR106		WSEQ_D	ATA106 [7:0)]			0000F000h
R12502 (30D6h)	WSEQ_Sequence_108		VSEQ_DEL	AY107 [3:0]	WS	EQ_DATA	START107	7 [3:0]		_ADDR107		WSEQ_D	ATA107 [7:0	0]			0000F000h
R12504 (30D8h)	WSEQ_Sequence_109	WSEQ_DA	_	-1108 [2:0] .AY108 [3:0]	WS	EQ_DATA_	START108	8 [3:0]		_ADDR108		WSEQ_D	ATA108 [7:0)]			0000F000h
R12506 (30DAh)	WSEQ_Sequence_110	V	_	AY109 [3:0]	WS	EQ_DATA	START109	9 [3:0]		_ADDR109		WSEQ_D	ATA109 [7:0)]			0000F000h
R12508 (30DCh)	WSEQ_Sequence_111		ATA_WIDTI VSEQ_DEL	-1110 [2:0] .AY110 [3:0]	WS	EQ_DATA	_START110	[3:0]	WSEC	_ADDR110	[12:0]	WSEQ_D	ATA110 [7:0)]			0000F000h
R12510 (30DEh)	WSEQ_Sequence_112	_	ATA_WIDTI VSEQ_DEL	H111 [2:0] .AY111 [3:0]]	WS	EQ_DATA	_START111	[3:0]	WSEC	_ADDR111	[12:0]	WSEQ_D	ATA111 [7:0]			02273A01h
R12512 (30E0h)	WSEQ_Sequence_113	WSEQ_D/	_	H112 [2:0] AY112 [3:0]]	WS	EQ_DATA	_START112	2 [3:0]	WSEC	_ADDR112	[12:0]	WSEQ_D	ATA112 [7:0)]			C2271300h
R12514 (30E2h)	WSEQ_Sequence_114	_	ATA_WIDTI VSEQ_DEL	H113 [2:0] AY113 [3:0]	WS	EQ_DATA	_START113	8 [3:0]	WSEG	_ADDR113	[12:0]	WSEQ_D	ATA113 [7:0)]			02270B00h
R12516 (30E4h)	WSEQ_Sequence_115		ATA_WIDTI VSEQ_DEL	H114 [2:0] AY114 [3:0]	WS	EQ_DATA	START114	[3:0]	WSEC	_ADDR114	[12:0]	WSEQ_D	ATA114 [7:0)]			0227FF01h
R12518 (30E6h)	WSEQ_Sequence_116	_	_	H115 [2:0] AY115 [3:0]	WS	EQ_DATA	START115	5 [3:0]	WSEC	_ADDR115	[12:0]	WSEQ_D	ATA115 [7:0)]			0000F000h
R12520 (30E8h)	WSEQ_Sequence_117	WSEQ_D/	ATA_WIDTI	-				START116		WSEC	_ADDR116	[12:0]		ATA116 [7:0				0000F000h
R12522 (30EAh)	WSEQ_Sequence_118		ATA_WIDTI VSEQ_DEL	H117 [2:0] AY117 [3:0]			START117		WSEC	_ADDR117	[12:0]		- ATA117 [7:0				0000F000h
R12524 (30ECh)	WSEQ_Sequence_119		ATA_WIDTI VSEQ_DEL	H118 [2:0] AY118 [3:0]	WS	EQ_DATA	START118	8 [3:0]	WSEC	_ADDR118	[12:0]	WSEQ_D	ATA118 [7:0)]			0000F000h
R12526 (30EEh)	WSEQ_Sequence_120	_	ATA_WIDTI VSEQ_DEL	H119 [2:0] AY119 [3:0]	WS	EQ_DATA	START119	9 [3:0]	WSEC	_ADDR119	[12:0]	WSEQ_D	ATA119 [7:0)]			0000F000h
R12528 (30F0h)	WSEQ_Sequence_121	_	ATA_WIDTH VSEQ_DEL	H120 [2:0] AY120 [3:0]	WS	EQ_DATA	START120) [3:0]	WSEC	_ADDR120	[12:0]	WSEQ_D	- ATA120 [7:0)]			0000F000h
R12530 (30F2h)	WSEQ_Sequence_122		ATA_WIDTH	H121 [2:0] AY121 [3:0]	WS	EQ_DATA	START121	[3:0]	WSEC	_ADDR121	[12:0]	WSEQ D	ATA121 [7:0	0]			0000F000h
R12532 (30F4h)	WSEQ_Sequence_123	WSEQ_D/	ATA_WIDTH					START122		WSEC	_ADDR122	[12:0]		ATA122 [7:0	-			0000F000h
R12534 (30F6h)	WSEQ_Sequence_124	WSEQ_D/	ATA_WIDTH					START123		WSEC	_ADDR123	[12:0]		ATA123 [7:0	-			0000F000h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12536 (30F8h)	WSEQ_Sequence_125	WSEQ_DAT	_	1124 [2:0] AY124 [3:0]		WS	EQ DATA	START124	[3:0]	WSEQ	_ADDR124	[12:0]	WSEQ DA	TA124 [7:0]		•	•	82283719h
R12538 (30FAh)	WSEQ_Sequence_126	WSEQ_DAT	TA_WIDTH		1			START125		WSEQ	_ADDR125	[12:0]		TA125 [7:0]				C2400001h
R12540 (30FCh)	WSEQ_Sequence_127	WSEQ_DAT	TA_WIDTH					START126		WSEQ	_ADDR126	[12:0]		TA126 [7:0]	-			02281301h
R12542 (30FEh)	WSEQ_Sequence_128	WSEQ_DAT	TA_WIDTH	1127 [2:0]						WSEQ	_ADDR127	[12:0]						8228191Fh
R12544 (3100h)	WSEQ_Sequence_129	WSEQ_DAT	ra_width					START127		WSEQ	_ADDR128	[12:0]		TA127 [7:0]	-			82410B02h
R12546 (3102h)	WSEQ_Sequence_130	WSEQ_DAT	TA_WIDTH					START128		WSEQ	_ADDR129	[12:0]		TA128 [7:0]				E241023Bh
R12548	WSEQ_Sequence_131	WSEQ_DAT	TA_WIDTH					START129		WSEQ	_ADDR130	[12:0]		TA129 [7:0]				02413B01h
(3104h) R12550	WSEQ_Sequence_132	WSEQ_DAT	_	AY130 [3:0] 1131 [2:0]		WSI	EQ_DATA_	START130	[3:0]	WSEQ	_ADDR131	[12:0]	WSEQ_DA	TA130 [7:0]]			62400000h
(3106h) R12552	WSEQ Sequence 133	WSEQ_DAT	_	AY131 [3:0] 1132 [2:0]		WS	EQ_DATA_	START131	[3:0]	WSEQ	ADDR132	[12:0]	WSEQ_DA	TA131 [7:0]]			E2414288h
(3108h) R12554	WSEQ_Sequence_134	WS WSEQ DAT	_	AY132 [3:0]		WSI	EQ_DATA_	START132	[3:0]	WSEC	ADDR133	[12:0]	WSEQ_DA	TA132 [7:0]]			02410B00h
(310Ah)			SEQ_DEL	AY133 [3:0]		WS	EQ_DATA_	START133	[3:0]		ADDR134		WSEQ_DA	TA133 [7:0]]			
R12556 (310Ch)	WSEQ_Sequence_135		SEQ_DEL	AY134 [3:0]		WSI	EQ_DATA_	START134	[3:0]				WSEQ_DA	TA134 [7:0]]			02410B00h
R12558 (310Eh)	WSEQ_Sequence_136		SEQ_DEL	AY135 [3:0]		WS	EQ_DATA_	START135	[3:0]		_ADDR135		WSEQ_DA	TA135 [7:0]]			02280E01h
R12560 (3110h)	WSEQ_Sequence_137	WSEQ_DAT	-	1136 [2:0] AY136 [3:0]		WS	EQ_DATA_	START136	[3:0]	WSEQ	_ADDR136	[12:0]	WSEQ_DA	TA136 [7:0]]			42410C03h
R12562 (3112h)	WSEQ_Sequence_138	WSEQ_DAT		1137 [2:0] AY137 [3:0]		WS	EQ DATA	START137	[3:0]	WSEQ	_ADDR137	[12:0]	WSEQ DA	TA137 [7:0]	1			E2410227h
R12564 (3114h)	WSEQ_Sequence_139	WSEQ_DAT	TA_WIDTH					START138		WSEQ	_ADDR138	[12:0]		TA138 [7:0]				02413B01h
R12566 (3116h)	WSEQ_Sequence_140	WSEQ_DAT	TA_WIDTH	1139 [2:0]						WSEQ	_ADDR139	[12:0]						E2414266h
R12568	WSEQ_Sequence_141	WSEQ_DAT	TA_WIDTH					START139		WSEQ	_ADDR140	[12:0]		TA139 [7:0]				E2415294h
(3118h) R12570	WSEQ_Sequence_142	WSEQ_DAT		AY140 [3:0] 1141 [2:0]		WSI	EQ_DATA_	START140	[3:0]	WSEQ	_ADDR141	[12:0]		TA140 [7:0]				02410B00h
(311Ah) R12572	WSEQ_Sequence_143	WSEQ_DAT	_	AY141 [3:0] 1142 [2:0]		WS	EQ_DATA_	START141	[3:0]	WSEQ	_ADDR142	[12:0]	WSEQ_DA	TA141 [7:0]]			02410B00h
(311Ch) R12574	WSEQ_Sequence_144	WSEQ_DAT	_	AY142 [3:0] 1143 [2:0]		WSI	EQ_DATA_	START142	[3:0]	WSEQ	ADDR143	[12:0]	WSEQ_DA	TA142 [7:0]				E2281734h
(311Eh) R12576	WSEQ Sequence 145	_	SEQ_DEL	AY143 [3:0]		WS	EQ_DATA_	START143	[3:0]		ADDR144		WSEQ_DA	TA143 [7:0]]			0228F501h
(3120h)	_ · _	_	_ SEQ_DEL	AY144 [3:0]		WS	EQ_DATA_	START144	[3:0]		ADDR145		WSEQ_DA	TA144 [7:0]]			
R12578 (3122h)	WSEQ_Sequence_146	 	_ SEQ_DEL	AY145 [3:0]		WSI	EQ_DATA_	START145	[3:0]		-		WSEQ_DA	TA145 [7:0]]			0000F000h
R12580 (3124h)	WSEQ_Sequence_147		SEQ_DEL	AY146 [3:0]		WSI	EQ_DATA_	START146	[3:0]	WSEQ	_ADDR146	[12:0]	WSEQ_DA	TA146 [7:0]]			0000F000h
R12582 (3126h)	WSEQ_Sequence_148	WSEQ_DAT		1147 [2:0] AY147 [3:0]		WSI	EQ_DATA_	START147	[3:0]	WSEQ	_ADDR147	[12:0]	WSEQ_DA	TA147 [7:0]]			0000F000h
R12584 (3128h)	WSEQ_Sequence_149	WSEQ_DAT	-	1148 [2:0] AY148 [3:0]		WS	EQ DATA	START148	[3:0]	WSEQ	_ADDR148	[12:0]	WSEQ DA	TA148 [7:0]	1			0000F000h
R12586 (312Ah)	WSEQ_Sequence_150	WSEQ_DAT	TA_WIDTH					START149		WSEQ	_ADDR149	[12:0]	_	TA149 [7:0]				0000F000h
R12588 (312Ch)	WSEQ_Sequence_151	WSEQ_DAT	ra_width	1150 [2:0]						WSEQ	_ADDR150	[12:0]	_	TA150 [7:0]				0000F000h
R12590	WSEQ_Sequence_152	WSEQ_DAT	TA_WIDTH					START150		WSEQ	_ADDR151	[12:0]	_					02283A01h
(312Eh) R12592	WSEQ_Sequence_153	WSEQ_DAT	TA_WIDTH					START151		WSEQ	_ADDR152	[12:0]	_	TA151 [7:0]				C2281300h
(3130h) R12594	WSEQ_Sequence_154	WSEQ_DAT		AY152 [3:0] 1153 [2:0]		WS	EQ_DATA_	START152	[3:0]	WSEQ	_ADDR153	[12:0]	WSEQ_DA	TA152 [7:0]]			02280B00h
(3132h) R12596	WSEQ Sequence 155	WS WSEQ_DAT		AY153 [3:0] 1154 [2:0]		WS	EQ_DATA_	START153	[3:0]	WSEQ	ADDR154	[12:0]	WSEQ_DA	TA153 [7:0]				0228FF01h
(3134h)	WSEQ Sequence 156		SEQ_DEL	AY154 [3:0]		WS	EQ_DATA_	START154	[3:0]		ADDR155		WSEQ_DA	TA154 [7:0]				0000F000h
(3136h)	_ · _	WS	SEQ_DEL	AY155 [3:0]		WS	EQ_DATA_	START155	[3:0]				WSEQ_DA	TA155 [7:0]				
(3138h)	WSEQ_Sequence_157		SEQ_DEL	AY156 [3:0]		WSI	EQ_DATA_	START156	[3:0]		_ADDR156		WSEQ_DA	TA156 [7:0]]			0000F000h
R12602 (313Ah)	WSEQ_Sequence_158	WSEQ_DAT		1157 [2:0] AY157 [3:0]		WS	EQ_DATA	START157	[3:0]	WSEQ	_ADDR157	[12:0]	WSEQ_DA	TA157 [7:0]]			0000F000h
R12604 (313Ch)	WSEQ_Sequence_159	WSEQ_DAT	TA_WIDTH					START158	-	WSEQ	_ADDR158	[12:0]		TA158 [7:0]	-			0000F000h
R12606 (313Eh)	WSEQ_Sequence_160	WSEQ_DAT	TA_WIDTH	1159 [2:0]						WSEQ	_ADDR159	[12:0]						0000F000h
		WS	SEQ_DEL	AY159 [3:0]		WS	EQ_DAIA_	START159	[3:0]				WSEQ_DA	TA159 [7:0]				I



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12608 (3140h)	WSEQ_Sequence_161	_	ATA_WIDTI WSEQ_DEI	H160 [2:0] AY160 [3:0]		WS	EQ_DATA_	START160	[3:0]	WSEC	_ADDR160	[12:0]	WSEQ_DA	ATA160 [7:0]				0000F000h
R12610 (3142h)	WSEQ_Sequence_162	_	ATA_WIDT	H161 [2:0] AY161 [3:0]		WSI	EQ_DATA_	START161	[3:0]	WSEC	_ADDR161	[12:0]	WSEQ_DA	ATA161 [7:0]]			0000F000h
R12612 (3144h)	WSEQ_Sequence_163	-	ATA_WIDTI WSEQ_DEI	H162 [2:0] .AY162 [3:0]		WS	EQ_DATA_	START162	[3:0]	WSEC	_ADDR162	[12:0]	WSEQ_DA	ATA162 [7:0]				0000F000h
R12614 (3146h)	WSEQ_Sequence_164	_	ATA_WIDTI NSEQ_DEI	H163 [2:0] AY163 [3:0]		WS	EQ_DATA_	START163	[3:0]	WSEC	_ADDR163	[12:0]	WSEQ_DA	ATA163 [7:0]]			0000F000h
R12616 (3148h)	WSEQ_Sequence_165	_	ATA_WIDTI NSEQ_DEL	H164 [2:0] .AY164 [3:0]		WS	EQ_DATA_	START164	[3:0]	WSEC	_ADDR164	[12:0]	WSEQ_DA	ATA164 [7:0]]			82293719h
R12618 (314Ah)	WSEQ_Sequence_166	_	ATA_WIDTI WSEQ_DEI	H165 [2:0] .AY165 [3:0]		WS	EQ_DATA_	START165	[3:0]	WSEC	ADDR165	[12:0]	WSEQ_DA	ATA165 [7:0]				C2500001h
R12620 (314Ch)	WSEQ_Sequence_167	_	ATA_WIDTI NSEQ_DEI	H166 [2:0] .AY166 [3:0]		WS	EQ_DATA_	START166	[3:0]	WSEC	_ADDR166	[12:0]	WSEQ_DA	ATA166 [7:0]]			02291301h
R12622 (314Eh)	WSEQ_Sequence_168	_	ata_widti NSEQ_dei	H167 [2:0] .AY167 [3:0]		WSI	EQ_DATA_	START167	[3:0]	WSEC	_ADDR167	[12:0]	WSEQ_DA	ATA167 [7:0]]			8229191Fh
R12624 (3150h)	WSEQ_Sequence_169		ATA_WIDTI WSEQ_DEI	H168 [2:0] .AY168 [3:0]		WS	EQ_DATA_	START168	[3:0]	WSEC	_ADDR168	[12:0]	WSEQ_DA	ATA168 [7:0]]			82510B00h
R12626 (3152h)	WSEQ_Sequence_170	_	ata_widti NSEQ_dei	H169 [2:0] .AY169 [3:0]		WSI	EQ_DATA_	START169	[3:0]	WSEC	_ADDR169	[12:0]	WSEQ_DA	ATA169 [7:0]]			E251023Bh
R12628 (3154h)	WSEQ_Sequence_171	_	ATA_WIDTI NSEQ_DEI	H170 [2:0] .AY170 [3:0]		WS	EQ_DATA_	START170	[3:0]	WSEC	_ADDR170	[12:0]	WSEQ_DA	ATA170 [7:0]]			02513B01h
R12630 (3156h)	WSEQ_Sequence_172	-	ATA_WIDTI WSEQ_DEI	H171 [2:0] .AY171 [3:0]		WS	EQ_DATA_	START171	[3:0]	WSEC	_ADDR171	[12:0]	WSEQ_DA	ATA171 [7:0]				62500000h
R12632 (3158h)	WSEQ_Sequence_173		ATA_WIDTI NSEQ_DEI	H172 [2:0] .AY172 [3:0]		WS	EQ_DATA_	START172	[3:0]	WSEC	_ADDR172	[12:0]	WSEQ_DA	ATA172 [7:0]]			E2514288h
R12634 (315Ah)	WSEQ_Sequence_174		ATA_WIDTI WSEQ_DEI	H173 [2:0] .AY173 [3:0]		WS	EQ_DATA_	START173	[3:0]	WSEC	_ADDR173	[12:0]	WSEQ_DA	ATA173 [7:0]]			02510B00h
R12636 (315Ch)	WSEQ_Sequence_175		ATA_WIDTI WSEQ_DEI	H174 [2:0] .AY174 [3:0]		WS	EQ_DATA_	START174	[3:0]	WSEC	_ADDR174	[12:0]	WSEQ_DA	ATA174 [7:0]]			02510B00h
R12638 (315Eh)	WSEQ_Sequence_176	_	ATA_WIDTI WSEQ_DEI	H175 [2:0] .AY175 [3:0]		WS	EQ_DATA_	START175	[3:0]	WSEC	_ADDR175	[12:0]	WSEQ_DA	ATA175 [7:0]]			02290E01h
R12640 (3160h)	WSEQ_Sequence_177		ATA_WIDTI WSEQ_DEI	H176 [2:0] .AY176 [3:0]		WS	EQ_DATA_	START176	[3:0]	WSEC	_ADDR176	[12:0]	WSEQ_DA	ATA176 [7:0]]			42510C02h
R12642 (3162h)	WSEQ_Sequence_178	_	ATA_WIDTI WSEQ_DEI	H177 [2:0] .AY177 [3:0]		WS	EQ_DATA_	START177	[3:0]	WSEC	_ADDR177	[12:0]	WSEQ_DA	ATA177 [7:0]]			E2510227h
R12644 (3164h)	WSEQ_Sequence_179		ATA_WIDTI WSEQ_DEI	H178 [2:0] .AY178 [3:0]		WS	EQ_DATA_	START178	[3:0]	WSEC	_ADDR178	[12:0]	WSEQ_DA	ATA178 [7:0]]			02513B01h
R12646 (3166h)	WSEQ_Sequence_180		ata_widti NSEQ_dei	H179 [2:0] .AY179 [3:0]		WSI	EQ_DATA_	START179	[3:0]	WSEC	_ADDR179	[12:0]	WSEQ_DA	ATA179 [7:0]]			E2514266h
R12648 (3168h)	WSEQ_Sequence_181	_	ATA_WIDTI NSEQ_DEI	H180 [2:0] .AY180 [3:0]		WS	EQ_DATA_	START180	[3:0]	WSEC	_ADDR180	[12:0]	WSEQ_DA	ATA180 [7:0]]			E2515294h
R12650 (316Ah)	WSEQ_Sequence_182	-	ata_widti NSEQ_dei	H181 [2:0] .AY181 [3:0]		WSI	EQ_DATA_	START181	[3:0]	WSEC	_ADDR181	[12:0]	WSEQ_DA	ATA181 [7:0]]			02510B00h
R12652 (316Ch)	WSEQ_Sequence_183			AY182 [3:0]		WS	EQ_DATA_	START182	[3:0]	WSEC	_ADDR182	[12:0]	WSEQ_DA	ATA182 [7:0]]			02510B00h
R12654 (316Eh)	WSEQ_Sequence_184	_	ata_widti NSEQ_dei	H183 [2:0] .AY183 [3:0]		WSI	EQ_DATA_	START183	[3:0]	WSEC	_ADDR183	[12:0]	WSEQ_DA	ATA183 [7:0]]			E2291734h
R12656 (3170h)	WSEQ_Sequence_185		ATA_WIDTI NSEQ_DEI	H184 [2:0] _AY184 [3:0]		WS	EQ_DATA_	START184	[3:0]	WSEC	_ADDR184	[12:0]	WSEQ_DA	ATA184 [7:0]]			0229F501h
R12658 (3172h)	WSEQ_Sequence_186	- 1		AY185 [3:0]		WS	EQ_DATA_	START185	[3:0]	WSEC	_ADDR185	[12:0]	WSEQ_DA	ATA185 [7:0]]			0000F000h
R12660 (3174h)	WSEQ_Sequence_187	/		AY186 [3:0]		WS	EQ_DATA_	START186	[3:0]		_ADDR186		WSEQ_DA	ATA186 [7:0]				0000F000h
R12662 (3176h)	WSEQ_Sequence_188	1		AY187 [3:0]		WS	EQ_DATA_	START187	[3:0]		_ADDR187		WSEQ_DA	ATA187 [7:0]				0000F000h
R12664 (3178h)	WSEQ_Sequence_189	1		AY188 [3:0]		WS	EQ_DATA_	START188	[3:0]		_ADDR188		WSEQ_DA	ATA188 [7:0]]			0000F000h
R12666 (317Ah)	WSEQ_Sequence_190	\		AY189 [3:0]		WS	EQ_DATA_	START189	[3:0]		_ADDR189		WSEQ_DA	ATA189 [7:0]]			0000F000h
R12668 (317Ch)	WSEQ_Sequence_191	- 1	NSEQ_DEL	H190 [2:0] .AY190 [3:0]		WS	EQ_DATA_	START190	[3:0]		ADDR190		WSEQ_DA	ATA190 [7:0]				0000F000h
(317Eh)	WSEQ_Sequence_192	- 1		AY191 [3:0]		WS	EQ_DATA_	START191	[3:0]		_ADDR191		WSEQ_DA	ATA191 [7:0]]			02293A01h
R12672 (3180h)	WSEQ_Sequence_193	- 1		AY192 [3:0]		WS	EQ_DATA_	START192	[3:0]		_ADDR192		WSEQ_DA	ATA192 [7:0]				C2291300h
R12674 (3182h)	WSEQ_Sequence_194	\	NSEQ_DEI	H193 [2:0] .AY193 [3:0]		WS	EQ_DATA_	START193	[3:0]	1	_ADDR193		WSEQ_DA	ATA193 [7:0]]			02290B00h
R12676 (3184h)	WSEQ_Sequence_195	- 1		AY194 [3:0]		WS	EQ_DATA_	START194	[3:0]		_ADDR194		WSEQ_DA	ATA194 [7:0]]			0229FF01h
R12678 (3186h)	WSEQ_Sequence_196	_	ATA_WIDTI WSEQ_DEI	H195 [2:0] .AY195 [3:0]		WS	EQ_DATA_	START195	[3:0]	WSEC	_ADDR195	[12:0]	WSEQ_DA	ATA195 [7:0]				0000F000h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12680 (3188h)	WSEQ_Sequence_197		ATA_WIDT NSEQ DEI	H196 [2:0] AY196 [3:0]		WS	EQ DATA	START196	[3:0]	WSEC	_ADDR196	[12:0]	WSEQ DA	ATA196 [7:0]	1	1		0000F000h
R12682 (318Ah)	WSEQ_Sequence_198	WSEQ_D	ATA_WIDT					START197		WSEC	_ADDR197	[12:0]		ATA197 [7:0]	-			0000F000h
R12684 (318Ch)	WSEQ_Sequence_199	-	ATA_WIDT	H198 [2:0] AY198 [3:0]				START198		WSEC	_ADDR198	[12:0]	WSEQ_DA	ATA198 [7:0]				0000F000h
R12686 (318Eh)	WSEQ_Sequence_200	_	ATA_WIDT NSEQ DEI	H199 [2:0] _AY199 [3:0		WS	eq data	START199	[3:0]	WSEC	_ADDR199	[12:0]	WSEQ DA	ATA199 [7:0]				0000F000h
R12688 (3190h)	WSEQ_Sequence_201	WSEQ_D	ATA_WIDT					START200		WSEC	_ADDR200	[12:0]		ATA200 [7:0]				0000F000h
R12690 (3192h)	WSEQ_Sequence_202	-	ATA_WIDT	H201 [2:0] AY201 [3:0]		WS	EQ_DATA	START201	[3:0]	WSEC	ADDR201	[12:0]	WSEQ_DA	ATA201 [7:0]				0000F000h
R12692 (3194h)	WSEQ_Sequence_203	_	ATA_WIDT NSEQ_DEI	H202 [2:0] .AY202 [3:0]		WS	EQ_DATA_	_START202	[3:0]	WSEC	_ADDR202	[12:0]	WSEQ_DA	ATA202 [7:0]]			0000F000h
R12694 (3196h)	WSEQ_Sequence_204	_	ATA_WIDT NSEQ_DEI	H203 [2:0] .AY203 [3:0]		WS	EQ_DATA_	_START203	[3:0]	WSEC	_ADDR203	[12:0]	WSEQ_DA	ATA203 [7:0]]			0000F000h
R12696 (3198h)	WSEQ_Sequence_205		ATA_WIDT NSEQ_DEI	H204 [2:0] .AY204 [3:0]		WS	EQ_DATA_	START204	[3:0]	WSEC	_ADDR204	[12:0]	WSEQ_DA	ATA204 [7:0]]			822A3719h
R12698 (319Ah)	WSEQ_Sequence_206		ATA_WIDT	H205 [2:0] AY205 [3:0]		WS	EQ_DATA	START205	[3:0]	WSEC	_ADDR205	[12:0]	WSEQ_DA	ATA205 [7:0]]			C2500001h
R12700 (319Ch)	WSEQ_Sequence_207	_	ATA_WIDT NSEQ_DEI	H206 [2:0] .AY206 [3:0]		WS	EQ_DATA_	START206	[3:0]	WSEC	_ADDR206	[12:0]	WSEQ_DA	ATA206 [7:0]]			022A1301h
R12702 (319Eh)	WSEQ_Sequence_208	_	ATA_WIDT	H207 [2:0] AY207 [3:0]		WS	EQ_DATA	START207	[3:0]	WSEC	_ADDR207	[12:0]	WSEQ_DA	ATA207 [7:0]]			822A191Fh
R12704 (31A0h)	WSEQ_Sequence_209		ATA_WIDT	H208 [2:0] AY208 [3:0				START208		WSEC	_ADDR208	[12:0]	_	ATA208 [7:0]				82510B02h
R12706 (31A2h)	WSEQ_Sequence_210	_	ATA_WIDT	H209 [2:0] AY209 [3:0]				START209		WSEC	_ADDR209	[12:0]	WSEQ_DA	ATA209 [7:0]				E251023Bh
R12708 (31A4h)	WSEQ_Sequence_211		ATA_WIDT	H210 [2:0] AY210 [3:0		WS	EQ DATA	START210	[3:0]	WSEC	_ADDR210	[12:0]	WSEQ DA	ATA210 [7:0]	1			02513B01h
R12710 (31A6h)	WSEQ_Sequence_212		ATA_WIDT	H211 [2:0] AY211 [3:0]				START211		WSEC	_ADDR211	[12:0]		ATA211 [7:0]	-			62500000h
R12712 (31A8h)	WSEQ_Sequence_213		ATA_WIDT	H212 [2:0] AY212 [3:0		WS	EQ DATA	START212	[3:0]	WSEC	_ADDR212	[12:0]		ATA212 [7:0]				E2514288h
R12714 (31AAh)	WSEQ_Sequence_214		ATA_WIDT NSEQ_DEI	H213 [2:0] AY213 [3:0]		WS	EQ_DATA_	START213	[3:0]	WSEC	_ADDR213	[12:0]	WSEQ_DA	ATA213 [7:0]]			02510B00h
R12716 (31ACh)	WSEQ_Sequence_215	-	ATA_WIDT NSEQ_DEI	H214 [2:0] .AY214 [3:0]		WS	EQ_DATA_	START214	[3:0]	WSEC	_ADDR214	[12:0]	WSEQ_DA	ATA214 [7:0]]			02510B00h
R12718 (31AEh)	WSEQ_Sequence_216		ATA_WIDT	H215 [2:0] AY215 [3:0]		WS	EQ_DATA	START215	[3:0]	WSEC	_ADDR215	[12:0]	WSEQ_DA	ATA215 [7:0]]			022A0E01h
R12720 (31B0h)	WSEQ_Sequence_217	_	ATA_WIDT	H216 [2:0] AY216 [3:0		WS	EQ_DATA	START216	[3:0]	WSEC	_ADDR216	[12:0]	WSEQ_DA	ATA216 [7:0]]			42510C03h
R12722 (31B2h)	WSEQ_Sequence_218		ATA_WIDT NSEQ_DEI	H217 [2:0] .AY217 [3:0]		WS	EQ_DATA_	START217	[3:0]	WSEC	_ADDR217	[12:0]	WSEQ_DA	ATA217 [7:0]]			E2510227h
R12724 (31B4h)	WSEQ_Sequence_219		ATA_WIDT	H218 [2:0] AY218 [3:0]		WS	EQ_DATA	START218	[3:0]	WSEC	_ADDR218	[12:0]	WSEQ_DA	ATA218 [7:0]]			02513B01h
R12726 (31B6h)	WSEQ_Sequence_220	_	ATA_WIDT NSEQ_DEI	H219 [2:0] .AY219 [3:0]		WS	EQ_DATA_	_START219	[3:0]	WSEC	_ADDR219	[12:0]	WSEQ_DA	ATA219 [7:0]]			E2514266h
R12728 (31B8h)	WSEQ_Sequence_221	WSEQ_D	ATA_WIDT					START220		WSEC	_ADDR220	[12:0]		ATA220 [7:0]				E2515294h
R12730 (31BAh)	WSEQ_Sequence_222		ATA_WIDT NSEQ_DEI	H221 [2:0] .AY221 [3:0]		WS	EQ_DATA_	START221	[3:0]	WSEC	_ADDR221	[12:0]	WSEQ_DA	ATA221 [7:0]]			02510B00h
R12732 (31BCh)	WSEQ_Sequence_223		ATA_WIDT NSEQ_DEI	H222 [2:0] .AY222 [3:0]		WS	EQ_DATA_	_START222	[3:0]	WSEC	_ADDR222	[12:0]	WSEQ_DA	ATA222 [7:0]]			02510B00h
R12734 (31BEh)	WSEQ_Sequence_224			H223 [2:0] .AY223 [3:0]		WS	EQ_DATA_	_START223	[3:0]	WSEC	_ADDR223	[12:0]	WSEQ_DA	ATA223 [7:0]]			E22A1734h
R12736 (31C0h)	WSEQ_Sequence_225		ATA_WIDT NSEQ_DEI	H224 [2:0] .AY224 [3:0]		WS	EQ_DATA_	_START224	[3:0]	WSEC	_ADDR224	[12:0]	WSEQ_DA	ATA224 [7:0]]			022AF501h
R12738 (31C2h)	WSEQ_Sequence_226		ATA_WIDT	H225 [2:0] AY225 [3:0]				START225		WSEC	_ADDR225	[12:0]	WSEQ_DA	ATA225 [7:0]]			0000F000h
R12740 (31C4h)	WSEQ_Sequence_227	_	_	H226 [2:0] AY226 [3:0				START226		WSEC	_ADDR226	[12:0]		ATA226 [7:0]	-			0000F000h
R12742 (31C6h)	WSEQ_Sequence_228	WSEQ_D	ATA_WIDT	H227 [2:0] AY227 [3:0]				START227		WSEC	_ADDR227	[12:0]		ATA227 [7:0]				0000F000h
R12744 (31C8h)	WSEQ_Sequence_229	WSEQ_D	ATA_WIDT	H228 [2:0] AY228 [3:0]				START228		WSEC	_ADDR228	[12:0]		ATA228 [7:0]	-			0000F000h
R12746 (31CAh)	WSEQ_Sequence_230	WSEQ_D	ATA_WIDT	H229 [2:0] AY229 [3:0				START229		WSEC	_ADDR229	[12:0]		ATA229 [7:0]				0000F000h
R12748 (31CCh)	WSEQ_Sequence_231	WSEQ_D	ATA_WIDT	H230 [2:0] AY230 [3:0				START230		WSEC	_ADDR230	[12:0]		ATA230 [7:0]				0000F000h
R12750 (31CEh)	WSEQ_Sequence_232	WSEQ_D	ATA_WIDT	H231 [2:0] AY231 [3:0				START231		WSEC	_ADDR231	[12:0]		ATA231 [7:0]				022A3A01h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12752 (31D0h)	WSEQ_Sequence_233		ATA_WIDTH	H232 [2:0] AY232 [3:0]	1	WS	EQ DATA	START232	[3:0]	WSEQ	_ADDR232	[12:0]	WSEQ D	ATA232 [7:0]				C22A1300h
R12754 (31D2h)	WSEQ_Sequence_234	WSEQ_D/	ATA_WIDTH					START233		WSEQ	_ADDR233	[12:0]		ATA233 [7:0]				022A0B00h
R12756 (31D4h)	WSEQ_Sequence_235		ATA_WIDTH VSEQ_DEL	H234 [2:0] AY234 [3:0]	WS	EQ_DATA_	_START234	[3:0]	WSEQ	_ADDR234	[12:0]	WSEQ_D	ATA234 [7:0]				022AFF01h
R12758 (31D6h)	WSEQ_Sequence_236	-	ATA_WIDTH	H235 [2:0] AY235 [3:0]	1	WS	έο σάτα	START235	[3:0]	WSEQ	_ADDR235	[12:0]	WSEO D	ATA235 [7:0]	1			0000F000h
R12760 (31D8h)	WSEQ_Sequence_237	WSEQ_D/	ATA_WIDTI					START236		WSEQ	_ADDR236	[12:0]		ATA236 [7:0]				0000F000h
R12762 (31DAh)	WSEQ_Sequence_238	-	ATA_WIDTH VSEQ_DEL	H237 [2:0] AY237 [3:0]	WS	EQ_DATA_	START237	[3:0]	WSEQ	_ADDR237	[12:0]	WSEQ_D/	ATA237 [7:0]				0000F000h
R12764 (31DCh)	WSEQ_Sequence_239	V	_	AY238 [3:0]	WS	EQ_DATA_	START238	[3:0]		_ADDR238		WSEQ_D/	ATA238 [7:0]				0000F000h
R12766 (31DEh)	WSEQ_Sequence_240	V	_	AY239 [3:0]	WS	EQ_DATA_	START239	[3:0]		_ADDR239		WSEQ_D/	ATA239 [7:0]				0000F000h
R12768 (31E0h) R12770	WSEQ_Sequence_241	V		AY240 [3:0]	WS	EQ_DATA_	START240	[3:0]		_ADDR240		WSEQ_D	ATA240 [7:0]				0000F000h
(31E2h) R12772	WSEQ_Sequence_242 WSEQ_Sequence_243	v	ATA_WIDTH VSEQ_DEL ATA_WIDTH	AY241 [3:0]	WS	eq_data_	START241	[3:0]		_ADDR241		WSEQ_D/	ATA241 [7:0]				0000F000h
(31E4h)	WSEQ_Sequence_243	v	_	AY242 [3:0]	WS	EQ_DATA_	START242	[3:0]		ADDR242		WSEQ_D/	ATA242 [7:0]				0000F000h
(31E6h) R12776	WSEQ_Sequence_244	v		AY243 [3:0]	WS	EQ_DATA_	START243	[3:0]		ADDR243		WSEQ_D/	ATA243 [7:0]				0000F000h
(31E8h) R12778	WSEQ Sequence 246	v	_	AY244 [3:0]	WS	EQ_DATA_	START244	[3:0]		ADDR245		WSEQ_D/	ATA244 [7:0]				0000F000h
(31EAh) R12780	WSEQ Sequence 247	V	_	AY245 [3:0]]	WS	EQ_DATA_	START245	[3:0]	WSEQ	ADDR246	[12:0]	WSEQ_D/	ATA245 [7:0]				0000F000h
(31ECh) R12782	WSEQ_Sequence_248	V	_	AY246 [3:0]]	WS	eq_data_	START246	[3:0]	WSEQ	_ _ADDR247	[12:0]	WSEQ_D/	ATA246 [7:0]				0000F000h
(31EEh) R12784	WSEQ_Sequence_249		VSEQ_DEL ATA_WIDTH	AY247 [3:0] H248 [2:0]]	WS	EQ_DATA_	START247	[3:0]	WSEQ	_ADDR248	[12:0]	WSEQ_D/	ATA247 [7:0]				0000F000h
(31F0h) R12786	WSEQ_Sequence_250		VSEQ_DEL ATA_WIDTH	AY248 [3:0 1249 [2:0]]	WS	eq_data_	START248	[3:0]	WSEQ	_ADDR249	[12:0]	WSEQ_D/	ATA248 [7:0]				0000F000h
(31F2h) R12788	WSEQ_Sequence_251	WSEQ_D/	ATA_WIDTH					START249		WSEQ	_ADDR250	[12:0]		ATA249 [7:0]				0000F000h
(31F4h) R12790	WSEQ_Sequence_252	WSEQ_D/	ATA_WIDTI					START250		WSEQ	_ADDR251	[12:0]		ATA250 [7:0]				0000F000h
(31F6h) R12792 (31F8h)	WSEQ_Sequence_253	WSEQ_D/	ATA_WIDTI		-			START251		WSEQ	_ADDR252	[12:0]	-	ATA251 [7:0] ATA252 [7:0]				0000F000h
(31FAh)	WSEQ_Sequence_254	WSEQ_D/	ATA_WIDTH	AY252 [3:0 1253 [2:0] AY253 [3:0	-			START252		WSEQ	_ADDR253	[12:0]	-	ATA252 [7:0] ATA253 [7:0]				0000F000h
R12796 (31FCh)	WSEQ_Sequence_255	WSEQ_D/	ATA_WIDTH		-			_START255		WSEQ	_ADDR254	[12:0]	-	ATA253 [7:0]				0000F000h
R12798 (31FEh)	WSEQ_Sequence_256	WSEQ_D/	ATA_WIDTH	-				START255		WSEQ	_ADDR255	[12:0]	-	ATA255 [7:0]				0000F000h
	WSEQ_Sequence_257	WSEQ_D/	ATA_WIDTH	-				START256		WSEQ	_ADDR256	[12:0]		ATA256 [7:0]				0000F000h
R12802 (3202h)	WSEQ_Sequence_258	WSEQ_D/	ATA_WIDTH					START257		WSEQ	_ADDR257	[12:0]	-	ATA257 [7:0]				0000F000h
R12804 (3204h)	WSEQ_Sequence_259		ATA_WIDTH VSEQ_DEL	H258 [2:0] AY258 [3:0]	WS	EQ_DATA_	START258	[3:0]	WSEQ	_ADDR258	[12:0]	WSEQ_D/	ATA258 [7:0]				0000F000h
R12806 (3206h)	WSEQ_Sequence_260		_	H259 [2:0] .AY259 [3:0]	WS	EQ_DATA_	START259	[3:0]	WSEQ	_ADDR259	[12:0]	WSEQ_D/	ATA259 [7:0]				0000F000h
R12808 (3208h)	WSEQ_Sequence_261	V	VSEQ_DEL	H260 [2:0] .AY260 [3:0]]	WS	EQ_DATA_	START260	[3:0]		_ADDR260		WSEQ_D/	ATA260 [7:0]				0000F000h
R12810 (320Ah)	WSEQ_Sequence_262	V	VSEQ_DEL	H261 [2:0] .AY261 [3:0]	WS	EQ_DATA_	START261	[3:0]		_ADDR261		WSEQ_D/	ATA261 [7:0]				0000F000h
(320Ch)	WSEQ_Sequence_263	V		AY262 [3:0]	WS	EQ_DATA_	START262	[3:0]		_ADDR262		WSEQ_D	ATA262 [7:0]				0000F000h
R12814 (320Eh)	WSEQ_Sequence_264	V	-	AY263 [3:0]	WS	EQ_DATA_	START263	[3:0]		_ADDR263		WSEQ_D/	ATA263 [7:0]				0000F000h
R12816 (3210h)	WSEQ_Sequence_265	V	_	AY264 [3:0]]	WS	EQ_DATA_	START264	[3:0]		_ADDR264		WSEQ_D/	ATA264 [7:0]				0000F000h
R12818 (3212h)	WSEQ_Sequence_266	V		AY265 [3:0]	WS	EQ_DATA_	START265	[3:0]	1	_ADDR265		WSEQ_D/	ATA265 [7:0]				0000F000h
R12820 (3214h)	WSEQ_Sequence_267	V		AY266 [3:0]	WS	EQ_DATA_	START266	[3:0]		_ADDR266		WSEQ_D/	ATA266 [7:0]				0000F000h
R12822 (3216h)	WSEQ_Sequence_268	-	_	H267 [2:0] .AY267 [3:0]	WS	EQ_DATA_	START267	[3:0]	WSEQ	_ADDR267	[12:0]	WSEQ_D/	ATA267 [7:0]				0000F000h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12824 (3218h)	WSEQ_Sequence_269		ATA_WIDT	H268 [2:0] LAY268 [3:0	1	WS	EQ DATA	START268	[3:0]	WSEQ	_ADDR268	[12:0]	WSEQ D	ATA268 [7:0	1			0000F000h
R12826 (321Ah)	WSEQ_Sequence_270	WSEQ_D	ATA_WIDT					START269		WSEQ	_ADDR269	[12:0]		ATA269 [7:0	-			0000F000h
R12828 (321Ch)	WSEQ_Sequence_271		ATA_WIDT	H270 [2:0] LAY270 [3:0				START270		WSEQ	_ADDR270	[12:0]		ATA270 [7:0	-			0000F000h
R12830 (321Eh)	WSEQ_Sequence_272		ATA_WIDT	H271 [2:0] LAY271 [3:0		WS	EQ DATA	START271	[3:0]	WSEQ	_ADDR271	[12:0]	WSEQ D	ATA271 [7:0	1			0000F000h
R12832 (3220h)	WSEQ_Sequence_273	WSEQ_D	ATA_WIDT					START272		WSEQ	_ADDR272	[12:0]	_	ATA272 [7:0	-			0000F000h
R12834 (3222h)	WSEQ_Sequence_274	WSEQ_D	ATA_WIDT	•				START273		WSEQ	_ADDR273	[12:0]		ATA273 [7:0				0000F000h
R12836 (3224h)	WSEQ_Sequence_275		ata_widt NSEQ dei	H274 [2:0] LAY274 [3:0				START274		WSEQ	_ADDR274	[12:0]		- ATA274 [7:0	-			0000F000h
R12838 (3226h)	WSEQ_Sequence_276	_	ata_widt //seq_dei	H275 [2:0] LAY275 [3:0		WS	eq data	START275	[3:0]	WSEQ	_ADDR275	[12:0]	WSEQ D	- ATA275 [7:0	1			0000F000h
R12840 (3228h)	WSEQ_Sequence_277		ata_widt NSEQ dei	H276 [2:0] LAY276 [3:0				START276		WSEQ	_ADDR276	[12:0]	WSEQ D	- ATA276 [7:0	1			0000F000h
R12842 (322Ah)	WSEQ_Sequence_278	WSEQ_D	ATA_WIDT					START277		WSEQ	_ADDR277	[12:0]	_	ATA277 [7:0				0000F000h
R12844 (322Ch)	WSEQ_Sequence_279	WSEQ_D	ATA_WIDT	-				- START278		WSEQ	_ADDR278	[12:0]		ATA278 [7:0	-			0000F000h
R12846 (322Eh)	WSEQ_Sequence_280	WSEQ_D	ATA_WIDT					START279		WSEQ	_ADDR279	[12:0]	_	ATA279 [7:0				0000F000h
R12848 (3230h)	WSEQ_Sequence_281	WSEQ_D	ATA_WIDT					START280		WSEQ	_ADDR280	[12:0]	_	ATA280 [7:0	-			0000F000h
R12850 (3232h)	WSEQ_Sequence_282	WSEQ_D	ATA_WIDT					START281		WSEQ	_ADDR281	[12:0]	_	ATA281 [7:0	-			0000F000h
R12852 (3234h)	WSEQ_Sequence_283	WSEQ_D	ATA_WIDT					START282		WSEQ	_ADDR282	[12:0]		ATA282 [7:0				0000F000h
R12854 (3236h)	WSEQ_Sequence_284	WSEQ_D	ATA_WIDT					START283		WSEQ	_ADDR283	[12:0]		ATA283 [7:0	-			0000F000h
R12856 (3238h)	WSEQ_Sequence_285	WSEQ_D	ATA_WIDT					START284		WSEQ	_ADDR284	[12:0]		ATA284 [7:0				0000F000h
R12858 (323Ah)	WSEQ_Sequence_286	WSEQ_D	ATA_WIDT					START285		WSEQ	_ADDR285	[12:0]		ATA285 [7:0	-			0000F000h
R12860 (323Ch)	WSEQ_Sequence_287	WSEQ_D	ATA_WIDT					START286		WSEQ	_ADDR286	[12:0]	-	ATA286 [7:0	-			0000F000h
R12862 (323Eh)	WSEQ_Sequence_288	WSEQ_D	ATA_WIDT					START287		WSEQ	_ADDR287	[12:0]		ATA287 [7:0	-			0000F000h
R12864 (3240h)	WSEQ_Sequence_289	WSEQ_D	ATA_WIDT	-				START288		WSEQ	_ADDR288	[12:0]		ATA288 [7:0	-			0000F000h
R12866 (3242h)	WSEQ_Sequence_290	WSEQ_D	ATA_WIDT			WS	eq data	START289	[3:0]	WSEQ	_ADDR289	[12:0]	-	ATA289 [7:0	-			0000F000h
R12868 (3244h)	WSEQ_Sequence_291	WSEQ_D	ATA_WIDT					- START290		WSEQ	_ADDR290	[12:0]	WSEQ D	ATA290 [7:0	1			0000F000h
R12870 (3246h)	WSEQ_Sequence_292	WSEQ_D	ATA_WIDT					START291		WSEQ	_ADDR291	[12:0]	-	ATA291 [7:0	-			0000F000h
	WSEQ_Sequence_293	WSEQ_D	ATA_WIDT	-				START292		WSEQ	_ADDR292	[12:0]		ATA292 [7:0				0000F000h
R12874 (324Ah)	WSEQ_Sequence_294	WSEQ_D	ATA_WIDT	-	-	WS	EQ DATA	- START293	[3:0]	WSEQ	_ADDR293	[12:0]	-	ATA293 [7:0	-			0000F000h
R12876 (324Ch)	WSEQ_Sequence_295	WSEQ_D	ATA_WIDT					START294		WSEQ	_ADDR294	[12:0]		ATA294 [7:0				0000F000h
R12878 (324Eh)	WSEQ_Sequence_296	WSEQ_D	ATA_WIDT	H295 [2:0] LAY295 [3:0	-			START295		WSEQ	_ADDR295	[12:0]	-	ATA295 [7:0	-			0000F000h
R12880 (3250h)	WSEQ_Sequence_297	WSEQ_D	ATA_WIDT	-				START296		WSEQ	_ADDR296	[12:0]		ATA296 [7:0	-			0000F000h
R12882 (3252h)	WSEQ_Sequence_298	WSEQ_D	ATA_WIDT	H297 [2:0] LAY297 [3:0	-			START297		WSEQ	_ADDR297	[12:0]		ATA297 [7:0				0000F000h
	WSEQ_Sequence_299	WSEQ_D	ATA_WIDT	H298 [2:0] LAY298 [3:0				START298		WSEQ	_ADDR298	[12:0]		ATA298 [7:0	-			0000F000h
	WSEQ_Sequence_300	WSEQ_D	ATA_WIDT	H299 [2:0] LAY299 [3:0						WSEQ	_ADDR299	[12:0]		ATA299 [7:0	-			0000F000h
R12888 (3258h)	WSEQ_Sequence_301	WSEQ_D	ATA_WIDT	H300 [2:0] LAY300 [3:0						WSEQ	_ADDR300	[12:0]		ATA300 [7:0	-			0000F000h
R12890 (325Ah)	WSEQ_Sequence_302	WSEQ_D	ATA_WIDT	H301 [2:0] LAY301 [3:0				_START301		WSEQ	_ADDR301	[12:0]		ATA301 [7:0	-			0000F000h
R12892 (325Ch)	WSEQ_Sequence_303	WSEQ_D	ATA_WIDT	H302 [2:0] LAY302 [3:0				_START301		WSEQ	_ADDR302	[12:0]		ATA302 [7:0	-			0000F000h
R12894 (325Eh)	WSEQ_Sequence_304	WSEQ_D	ATA_WIDT	H303 [2:0] LAY303 [3:0				_START302		WSEQ	_ADDR303	[12:0]		ATA302 [7:0				0000F000h
(020211)	1	1 1	NOEQ_DEI	LAI 303 [3:0	1	۷۷5		_JIAR1303	[3.0]	1			woev_D/	-17303[7:0	1			1



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12896 (3260h)	WSEQ_Sequence_305		ATA_WIDTI VSEQ_DEI	H304 [2:0] .AY304 [3:0	1	WS	EQ DATA	START304	[3:0]	WSEQ	_ADDR304	[12:0]	WSEQ D	ATA304 [7:0	1			0000F000h
R12898 (3262h)	WSEQ_Sequence_306	WSEQ_D	ATA_WIDTI					START305		WSEQ	_ADDR305	[12:0]		ATA305 [7:0]				0000F000h
R12900 (3264h)	WSEQ_Sequence_307		ATA_WIDTI VSEQ_DEI	H306 [2:0] AY306 [3:0]	WS	EQ_DATA_	START306	[3:0]	WSEQ	_ADDR306	[12:0]	WSEQ_D	ATA306 [7:0]]			0000F000h
R12902 (3266h)	WSEQ_Sequence_308	-	ATA_WIDTI	H307 [2:0] AY307 [3:0]	1	WS	FQ DATA	START307	' [3·0]	WSEQ	_ADDR307	[12:0]	WSEQ D	ATA307 [7:0	1			0000F000h
R12904 (3268h)	WSEQ_Sequence_309	WSEQ_D	ATA_WIDTI					START308		WSEQ	_ADDR308	[12:0]		ATA308 [7:0]				0000F000h
R12906 (326Ah)	WSEQ_Sequence_310	_	ATA_WIDTI NSEQ_DEI	H309 [2:0] AY309 [3:0]	WS	EQ_DATA	START309	[3:0]	WSEQ	_ADDR309	[12:0]	WSEQ_D	ATA309 [7:0]]			0000F000h
R12908 (326Ch)	WSEQ_Sequence_311	1		AY310 [3:0]]	WS	EQ_DATA_	START310	[3:0]		_ADDR310		WSEQ_D	ATA310 [7:0]			0000F000h
R12910 (326Eh)	WSEQ_Sequence_312			AY311 [3:0]		WS	EQ_DATA_	START311	[3:0]		_ADDR311		WSEQ_D	ATA311 [7:0]				0000F000h
R12912 (3270h) R12914	WSEQ_Sequence_313		_	AY312 [3:0]	WS	EQ_DATA_	START312	[3:0]		ADDR312		WSEQ_D	ATA312 [7:0]]			0000F000h
(3272h) R12916	WSEQ_Sequence_314 WSEQ_Sequence_315		ATA_WIDTI VSEQ_DEL ATA_WIDTI	AY313 [3:0]		WS	EQ_DATA_	START313	[3:0]		_ADDR313 ADDR314		WSEQ_D	ATA313 [7:0]				0000F000h
(3274h) R12918	WSEQ_Sequence_315		_	AY314 [3:0]	WS	EQ_DATA_	START314	[3:0]		_ADDR314		WSEQ_D	ATA314 [7:0]]			0000F000h
(3276h) R12920	WSEQ_Sequence_317			AY315 [3:0]	WS	EQ_DATA_	START315	[3:0]		ADDR318		WSEQ_D	ATA315 [7:0]]			0000F000h
(3278h) R12922	WSEQ_Sequence_318		_	AY316 [3:0]]	WS	EQ_DATA_	START316	[3:0]		_ADDR317		WSEQ_D	ATA316 [7:0]]			0000F000h
(327Ah) R12924	WSEQ Sequence 319	1		AY317 [3:0]]	WS	EQ_DATA_	START317	[3:0]		- ADDR318		WSEQ_D	ATA317 [7:0]				0000F000h
(327Ch) R12926	WSEQ_Sequence_320		- VSEQ_DEL ATA_WIDTI	AY318 [3:0 H319 [2:0]]	WS	EQ_DATA_	START318	[3:0]	WSEC	ADDR319	[12:0]	WSEQ_D	ATA318 [7:0]]			0000F000h
(327Eh) R12928	WSEQ_Sequence_321		NSEQ_DEL ATA_WIDTI	AY319 [3:0 H320 [2:0]		WS	EQ_DATA_	START319	[3:0]	WSEQ	_ADDR320	[12:0]	WSEQ_D	ATA319 [7:0]				0000F000h
(3280h) R12930	WSEQ_Sequence_322		VSEQ_DEL ATA_WIDTI	.AY320 [3:0] H321 [2:0]		WS	EQ_DATA_	START320	[3:0]	WSEQ	_ADDR321	[12:0]	WSEQ_D	ATA320 [7:0]				0000F000h
(3282h) R12932	WSEQ_Sequence_323	WSEQ_D	ATA_WIDTI					START321		WSEQ	_ADDR322	[12:0]		ATA321 [7:0]				0000F000h
(3284h) R12934 (2286b)	WSEQ_Sequence_324	WSEQ_D	ATA_WIDTI					START322		WSEQ	_ADDR323	[12:0]		ATA322 [7:0]				0000F000h
(3286h) R12936 (3288h)	WSEQ_Sequence_325	WSEQ_D	ATA_WIDTI					START323		WSEQ	_ADDR324	[12:0]	_	ATA323 [7:0]				0000F000h
R12938 (328Ah)	WSEQ_Sequence_326	WSEQ_D	ATA_WIDT	AY324 [3:0] H325 [2:0] AY325 [3:0]				_START324		WSEQ	_ADDR325	[12:0]	-	ATA324 [7:0]				0000F000h
R12940 (328Ch)	WSEQ_Sequence_327	WSEQ_D	ATA_WIDTI	H326 [2:0]				-		WSEQ	_ADDR326	[12:0]						0000F000h
(328Ch) (328Eh)	WSEQ_Sequence_328	WSEQ_D	ATA_WIDT					START326		WSEQ	_ADDR327	[12:0]	_	ATA326 [7:0]				0000F000h
(320L1) R12944 (3290h)	WSEQ_Sequence_329	WSEQ_D	ATA_WIDTI	AY327 [3:0] H328 [2:0] AY328 [3:0]				START327		WSEQ	_ADDR328	[12:0]		ATA327 [7:0] ATA328 [7:0]				0000F000h
R12946 (3292h)	WSEQ_Sequence_330	WSEQ_D	ATA_WIDTI					START320		WSEQ	_ADDR329	[12:0]	_	ATA328 [7:0]				0000F000h
	WSEQ_Sequence_331	WSEQ_D	ATA_WIDTI							WSEQ	_ADDR330	[12:0]	_	ATA330 [7:0]				0000F000h
R12950 (3296h)	WSEQ_Sequence_332	WSEQ_D	ATA_WIDT	H331 [2:0] .AY331 [3:0				START331		WSEQ	_ADDR331	[12:0]	-	ATA331 [7:0				0000F000h
R12952 (3298h)	WSEQ_Sequence_333	WSEQ_D	ATA_WIDTI	H332 [2:0] .AY332 [3:0				START332		WSEQ	_ADDR332	[12:0]		ATA332 [7:0]				0000F000h
R12954 (329Ah)	WSEQ_Sequence_334		VSEQ_DEI	H333 [2:0] AY333 [3:0]	WS	EQ_DATA_	START333	[3:0]		_ADDR333		WSEQ_D	ATA333 [7:0]			0000F000h
R12956 (329Ch)	WSEQ_Sequence_335	- 1		AY334 [3:0]	WS	EQ_DATA_	START334	[3:0]		_ADDR334		WSEQ_D	ATA334 [7:0]]			0000F000h
R12958 (329Eh)	WSEQ_Sequence_336	- 1	VSEQ_DEI	H335 [2:0] .AY335 [3:0]]	WS	EQ_DATA_	START335	[3:0]		_ADDR335		WSEQ_D	ATA335 [7:0]]			0000F000h
R12960 (32A0h)	WSEQ_Sequence_337	- \	_	AY336 [3:0]	WS	EQ_DATA_	START336	[3:0]		_ADDR336		WSEQ_D	ATA336 [7:0]			0000F000h
R12962 (32A2h)	WSEQ_Sequence_338		ATA_WIDTI	H337 [2:0] _AY337 [3:0]]	WS	EQ_DATA	START337	[3:0]	WSEQ	_ADDR337	[12:0]	WSEQ_D	ATA337 [7:0]]			0000F000h
R12964 (32A4h)	WSEQ_Sequence_339	WSEQ_D	ATA_WIDTI	-	-			START338		WSEQ	_ADDR338	[12:0]		ATA338 [7:0	-			0000F000h
R12966 (32A6h)	WSEQ_Sequence_340	WSEQ_D	ATA_WIDTI	H339 [2:0] AY339 [3:0				START339		WSEQ	_ADDR339	[12:0]		ATA339 [7:0]				0000F000h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R12968 (32A8h)	WSEQ_Sequence_341		ATA_WIDTI VSEQ DEI	H340 [2:0] .AY340 [3:0	1	WS	EQ DATA	START340	[3:0]	WSEQ	_ADDR340	[12:0]	WSEQ DA	ATA340 [7:0	1	1		0000F000h
R12970 (32AAh)	WSEQ_Sequence_342	WSEQ_D	ATA_WIDT	-				START341		WSEQ	_ADDR341	[12:0]	WSEQ DA	- ATA341 [7:0	1			0000F000h
R12972 (32ACh)	WSEQ_Sequence_343	_	ATA_WIDTI VSEQ_DEI	H342 [2:0] .AY342 [3:0	1			START342		WSEQ	_ADDR342	[12:0]		ATA342 [7:0	-			0000F000h
R12974 (32AEh)	WSEQ_Sequence_344		ATA_WIDTI	H343 [2:0] .AY343 [3:0	1	WS	EQ DATA	START343	[3:0]	WSEQ	_ADDR343	[12:0]	WSEQ DA	ATA343 [7:0	1			0000F000h
R12976 (32B0h)	WSEQ_Sequence_345	WSEQ_D	ATA_WIDTI					START344		WSEQ	_ADDR344	[12:0]		ATA344 [7:0	-			0000F000h
R12978 (32B2h)	WSEQ_Sequence_346	-	ATA_WIDTI VSEQ_DEI	H345 [2:0] .AY345 [3:0	1	WS	eq data	START345	[3:0]	WSEQ	_ADDR345	[12:0]	WSEQ DA	ATA345 [7:0	1			0000F000h
R12980 (32B4h)	WSEQ_Sequence_347	_	ATA_WIDTI VSEQ_DEL	H346 [2:0] .AY346 [3:0	1	WS	EQ DATA	START346	[3:0]	WSEQ	_ADDR346	[12:0]		ATA346 [7:0				0000F000h
R12982 (32B6h)	WSEQ_Sequence_348		ATA_WIDTI NSEQ_DEI	H347 [2:0] .AY347 [3:0]	WS	EQ_DATA_	START347	[3:0]	WSEQ	_ADDR347	[12:0]	WSEQ_DA	ATA347 [7:0]			0000F000h
R12984 (32B8h)	WSEQ_Sequence_349		ATA_WIDTI NSEQ_DEI	H348 [2:0] .AY348 [3:0]	WS	EQ_DATA_	START348	[3:0]	WSEQ	_ADDR348	[12:0]	WSEQ_DA	ATA348 [7:0]			0000F000h
R12986 (32BAh)	WSEQ_Sequence_350		ATA_WIDTI NSEQ_DEI	H349 [2:0] .AY349 [3:0]	WS	EQ_DATA_	START349	[3:0]	WSEQ	_ADDR349	[12:0]	WSEQ_DA	ATA349 [7:0]			0000F000h
R12988 (32BCh)	WSEQ_Sequence_351		ATA_WIDTI NSEQ_DEI	H350 [2:0] .AY350 [3:0]	WS	EQ_DATA_	START350	[3:0]	WSEQ	_ADDR350	[12:0]	WSEQ_DA	ATA350 [7:0]			0000F000h
R12990 (32BEh)	WSEQ_Sequence_352	_	ATA_WIDTI NSEQ_DEI	H351 [2:0] .AY351 [3:0]	WS	EQ_DATA_	START351	[3:0]	WSEQ	_ADDR351	[12:0]	WSEQ_DA	ATA351 [7:0]			0000F000h
R12992 (32C0h)	WSEQ_Sequence_353	-	ATA_WIDTI NSEQ_DEI	H352 [2:0] .AY352 [3:0]	WS	EQ_DATA_	START352	[3:0]	WSEQ	_ADDR352	[12:0]	WSEQ_DA	ATA352 [7:0]			0000F000h
R12994 (32C2h)	WSEQ_Sequence_354	-	ATA_WIDTI VSEQ_DEI	H353 [2:0] .AY353 [3:0]	WS	EQ_DATA_	START353	[3:0]	WSEQ	_ADDR353	[12:0]	WSEQ_DA	ATA353 [7:0]			0000F000h
R12996 (32C4h)	WSEQ_Sequence_355		ATA_WIDTI VSEQ_DEI	H354 [2:0] .AY354 [3:0]	WS	EQ_DATA_	START354	[3:0]	WSEQ	_ADDR354	[12:0]	WSEQ_DA	ATA354 [7:0]			0000F000h
R12998 (32C6h)	WSEQ_Sequence_356		ATA_WIDTI NSEQ_DEI	H355 [2:0] .AY355 [3:0]	WS	EQ_DATA_	START355	[3:0]	WSEQ	_ADDR355	[12:0]	WSEQ_DA	ATA355 [7:0]			0000F000h
R13000 (32C8h)	WSEQ_Sequence_357		ATA_WIDTI VSEQ_DEI	H356 [2:0] .AY356 [3:0]	WS	EQ_DATA_	START356	[3:0]	WSEQ	_ADDR356	[12:0]	WSEQ_DA	ATA356 [7:0]			0000F000h
R13002 (32CAh)	WSEQ_Sequence_358	-	ATA_WIDTI NSEQ_DEI	H357 [2:0] .AY357 [3:0]	WS	EQ_DATA_	START357	[3:0]	WSEQ	_ADDR357	[12:0]	WSEQ_DA	ATA357 [7:0]			0000F000h
R13004 (32CCh)	WSEQ_Sequence_359			AY358 [3:0]	WS	EQ_DATA_	START358	[3:0]	WSEQ	_ADDR358	[12:0]	WSEQ_DA	ATA358 [7:0]			0000F000h
R13006 (32CEh)	WSEQ_Sequence_360	- \	_	AY359 [3:0]	WS	EQ_DATA_	START359	[3:0]		_ADDR359		WSEQ_DA	ATA359 [7:0]			0000F000h
R13008 (32D0h)	WSEQ_Sequence_361		_	AY360 [3:0]	WS	EQ_DATA_	START360	[3:0]		_ADDR360		WSEQ_DA	ATA360 [7:0]			0000F000h
R13010 (32D2h)	WSEQ_Sequence_362		_	AY361 [3:0]	WS	EQ_DATA_	START361	[3:0]		_ADDR361		WSEQ_DA	ATA361 [7:0]			0000F000h
R13012 (32D4h)	WSEQ_Sequence_363		_	AY362 [3:0]	WS	EQ_DATA_	START362	[3:0]		_ADDR362		WSEQ_DA	ATA362 [7:0]			0000F000h
R13014 (32D6h)	WSEQ_Sequence_364	- \		AY363 [3:0]	WS	EQ_DATA_	START363	[3:0]		_ADDR363		WSEQ_DA	ATA363 [7:0]			0000F000h
(32D8h)	WSEQ_Sequence_365			AY364 [3:0]	WS	EQ_DATA_	START364	[3:0]		_ADDR364		WSEQ_DA	ATA364 [7:0]			0000F000h
R13018 (32DAh)	WSEQ_Sequence_366		_	AY365 [3:0]	WS	EQ_DATA_	START365	[3:0]		_ADDR365		WSEQ_DA	ATA365 [7:0]			0000F000h
R13020 (32DCh)	WSEQ_Sequence_367	1		AY366 [3:0]	WS	EQ_DATA_	START366	[3:0]		_ADDR366		WSEQ_DA	ATA366 [7:0]			0000F000h
R13022 (32DEh)	WSEQ_Sequence_368	- 1	VSEQ_DEI	H367 [2:0] .AY367 [3:0]	WS	EQ_DATA_	START367	[3:0]		_ADDR367		WSEQ_DA	ATA367 [7:0]			0000F000h
R13024 (32E0h)	WSEQ_Sequence_369	1	VSEQ_DEL	H368 [2:0] _AY368 [3:0]	WS	EQ_DATA_	START368	[3:0]		_ADDR368		WSEQ_DA	ATA368 [7:0]			0000F000h
(32E2h)	WSEQ_Sequence_370	- 1	VSEQ_DEL	H369 [2:0] _AY369 [3:0]	WS	EQ_DATA_	START369	[3:0]		_ADDR369		WSEQ_DA	ATA369 [7:0]			0000F000h
R13028 (32E4h)	WSEQ_Sequence_371	- \	VSEQ_DEL	H370 [2:0] .AY370 [3:0]	WS	EQ_DATA_	START370	[3:0]		_ADDR370		WSEQ_DA	ATA370 [7:0]			0000F000h
R13030 (32E6h)	WSEQ_Sequence_372		VSEQ_DEI	H371 [2:0] .AY371 [3:0]	WS	EQ_DATA_	START371	[3:0]		_ADDR371		WSEQ_DA	ATA371 [7:0]			0000F000h
R13032 (32E8h)	WSEQ_Sequence_373	- \	VSEQ_DEL	H372 [2:0] .AY372 [3:0]	WS	EQ_DATA_	START372	[3:0]		_ADDR372		WSEQ_DA	ATA372 [7:0]			0000F000h
R13034 (32EAh)	WSEQ_Sequence_374			AY373 [3:0]	WS	EQ_DATA_	START373	[3:0]		_ADDR373		WSEQ_DA	ATA373 [7:0]			0000F000h
R13036 (32ECh)	WSEQ_Sequence_375	- \	_	AY374 [3:0]	WS	EQ_DATA_	START374	[3:0]		_ADDR374		WSEQ_DA	ATA374 [7:0]			0000F000h
R13038 (32EEh)	WSEQ_Sequence_376		ATA_WIDTI NSEQ_DEI	H375 [2:0] _AY375 [3:0]	WS	EQ_DATA_	START375	[3:0]	WSEQ	_ADDR375	[12:0]	WSEQ_DA	ATA375 [7:0]			0000F000h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R13040 (32F0h)	WSEQ_Sequence_377		ATA_WIDTH	H376 [2:0] .AY376 [3:0	1	WS	EQ_DATA	START376	6 [3:0]	WSEC	_ADDR376	6 [12:0]	WSEQ D	ATA376 [7:0				0000F000h
R13042 (32F2h)	WSEQ_Sequence_378	WSEQ_D	ATA_WIDTI				EQ DATA	-		WSEC	_ADDR377	7 [12:0]		ATA377 [7:0				0000F000h
R13044 (32F4h)	WSEQ_Sequence_379	WSEQ_D	ATA_WIDTI				EQ_DATA	-		WSEC	_ADDR378	8 [12:0]		ATA378 [7:0				0000F000h
R13046 (32F6h)	WSEQ_Sequence_380			H379 [2:0] AY379 [3:0	1	We		STADT27	0 [3:0]	WSEC	_ADDR379	9 [12:0]		ATA379 [7:0	1			0000F000h
R13048 (32F8h)	WSEQ_Sequence_381	WSEQ_D	ATA_WIDTI				eq_data_	-		WSEC	_ADDR380	0 [12:0]		ATA379 [7:0				0000F000h
R13050 (32FAh)	WSEQ_Sequence_382	WSEQ_D	ATA_WIDTH				EQ DATA	-		WSEC	_ADDR381	[12:0]	_	ATA381 [7:0				0000F000h
R13052 (32FCh)	WSEQ_Sequence_383	WSEQ_D	ATA_WIDTH		-	WS	EQ_DATA_	START382	2 [3:0]	WSEC	_ADDR382	2 [12:0]		ATA382 [7:0	-			0000F000h
R13054 (32FEh)	WSEQ_Sequence_384		_	AY383 [3:0]	WS	EQ_DATA_	START38	3 [3:0]		_ADDR383		WSEQ_D/	ATA383 [7:0)]			0000F000h
R13056 (3300h)	WSEQ_Sequence_385			AY384 [3:0]	WS	EQ_DATA_	_START384	4 [3:0]		_ADDR384		WSEQ_D/	ATA384 [7:0]			FFFFFFF
R13058 (3302h)	WSEQ_Sequence_386		-	AY385 [3:0]	WS	EQ_DATA_	_START38	5 [3:0]		ADDR385		WSEQ_D/	ATA385 [7:0]			FFFFFFF
R13060 (3304h)	WSEQ_Sequence_387		ata_width VSEQ_del Ata_width	AY386 [3:0]	WS	EQ_DATA_	_START386	6 [3:0]		ADDR386		WSEQ_D	ATA386 [7:0]			FFFFFFFh
R13062 (3306h) R13064	WSEQ_Sequence_388 WSEQ_Sequence_389		_	AY387 [3:0]	WS	EQ_DATA_	START387	7 [3:0]		ADDR387		WSEQ_D/	ATA387 [7:0]			FFFFFFFh
(3308h) R13066	WSEQ_Sequence_390		_	AY388 [3:0]	WS	EQ_DATA_	_START388	8 [3:0]		ADDR38		WSEQ_D/	ATA388 [7:0]			FFFFFFF
(330Ah) R13068	WSEQ_Sequence 391	- 1	-	AY389 [3:0]]	WS	EQ_DATA_	START38	9 [3:0]		ADDR390		WSEQ_D/	ATA389 [7:0)]			FFFFFFF
(330Ch) R13070	WSEQ_Sequence_392		_	AY390 [3:0]]	WS	EQ_DATA_	_START390	0 [3:0]		ADDR391		WSEQ_D/	ATA390 [7:0]			FFFFFFF
(330Eh) R13072	WSEQ_Sequence_393		VSEQ_DEL ATA_WIDTH	AY391 [3:0 H392 [2:0]]	WS	EQ_DATA_	_START39	1 [3:0]	WSEC	_ADDR392	2 [12:0]	WSEQ_D/	ATA391 [7:0]			FFFFFFFh
(3310h) R13074	WSEQ_Sequence_394		VSEQ_DEL ATA_WIDTH	.AY392 [3:0] H393 [2:0]]	WS	EQ_DATA_	_START392	2 [3:0]	WSEC	_ADDR393	8 [12:0]	WSEQ_D	ATA392 [7:0]			FFFFFFFh
(3312h) R13076	WSEQ_Sequence_395	WSEQ_D	ATA_WIDTH				EQ_DATA_	-		WSEC	_ADDR394	l [12:0]		ATA393 [7:0				FFFFFFF
(3314h) R13078	WSEQ_Sequence_396	WSEQ_D	ATA_WIDTI				EQ_DATA_	-		WSEC	_ADDR395	5 [12:0]		ATA394 [7:0				FFFFFFFh
(3316h) R13080 (3318h)	WSEQ_Sequence_397	WSEQ_D	ATA_WIDTI	AY395 [3:0] H396 [2:0] AY396 [3:0]	-		eq_data_			WSEC	_ADDR396	6 [12:0]		ATA395 [7:0 ATA396 [7:0				FFFFFFF
R13082 (331Ah)	WSEQ_Sequence_398	WSEQ_D	ATA_WIDTH		-		EQ_DATA	-		WSEC	_ADDR397	7 [12:0]		ATA390 [7:0				FFFFFFF
R13084 (331Ch)	WSEQ_Sequence_399	WSEQ_D	ATA_WIDTH		-		EQ DATA			WSEC	_ADDR398	8 [12:0]		ATA398 [7:0				FFFFFFF
R13086 (331Eh)	WSEQ_Sequence_400	WSEQ_D	ATA_WIDTH				EQ_DATA	-		WSEC	_ADDR399	9 [12:0]	_	ATA399 [7:0				FFFFFFF
R13088 (3320h)	WSEQ_Sequence_401	WSEQ_D	ATA_WIDTH	-			EQ_DATA			WSEC	_ADDR400) [12:0]		ATA400 [7:0				FFFFFFFh
R13090 (3322h)	WSEQ_Sequence_402		ATA_WIDTH VSEQ_DEL	H401 [2:0] AY401 [3:0]]	WS	EQ_DATA_	_START40 ⁴	1 [3:0]	WSEC	_ADDR401	[12:0]	WSEQ_D	ATA401 [7:0]			FFFFFFFh
R13092 (3324h)	WSEQ_Sequence_403			AY402 [3:0]	WS	EQ_DATA_	_START402	2 [3:0]	WSEC	_ADDR402	2 [12:0]	WSEQ_D/	ATA402 [7:0]			FFFFFFFh
R13094 (3326h)	WSEQ_Sequence_404		VSEQ_DEL	H403 [2:0] .AY403 [3:0]	WS	EQ_DATA_	_START403	3 [3:0]	WSEC	_ADDR403	8 [12:0]	WSEQ_D/	ATA403 [7:0]			FFFFFFFh
R13096 (3328h)	WSEQ_Sequence_405	\	_	AY404 [3:0]]	WS	EQ_DATA_	START404	4 [3:0]		_ADDR404		WSEQ_D/	ATA404 [7:0]			FFFFFFFh
R13098 (332Ah)	WSEQ_Sequence_406			AY405 [3:0]]	WS	EQ_DATA_	_START40	5 [3:0]		ADDR40		WSEQ_D/	ATA405 [7:0]			FFFFFFFh
R13100 (332Ch)	WSEQ_Sequence_407			AY406 [3:0]	WS	EQ_DATA_	_START406	6 [3:0]		ADDR406		WSEQ_D/	ATA406 [7:0]			FFFFFFF
R13102 (332Eh) R13104	WSEQ_Sequence_408		ATA_WIDTH VSEQ_DEL ATA_WIDTH	AY407 [3:0]	WS	EQ_DATA_	START407	7 [3:0]		ADDR407		WSEQ_D/	ATA407 [7:0]			FFFFFFFh
(3330h) R13106	WSEQ_Sequence_409 WSEQ_Sequence_410		_	AY408 [3:0]	WS	EQ_DATA_	_START408	8 [3:0]		ADDR408		WSEQ_D/	ATA408 [7:0]			FFFFFFF
(3332h) R13108	WSEQ_Sequence_410	\	_	AY409 [3:0]	WS	EQ_DATA_	_START409	9 [3:0]		ADDR40		WSEQ_D/	ATA409 [7:0]			FFFFFFF
(3334h) R13110	WSEQ_Sequence_411	1	_	AY410 [3:0]]	WS	EQ_DATA_	START410	0 [3:0]		ADDR411		WSEQ_D/	ATA410 [7:0)]			FFFFFFF
(3336h)		_	_	_AY411 [3:0]]	WS	EQ_DATA_	_START411	1 [3:0]			·•]	WSEQ_D	ATA411 [7:0]			



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R13112 (3338h)	WSEQ_Sequence_413	WSEQ_D	ATA_WIDTI					START412	[3:0]	WSEQ	_ADDR412	-	WSEO D	TA412 [7:0]			1	FFFFFFFh
R13114 (333Ah)	WSEQ_Sequence_414	WSEQ_D	ATA_WIDTI	H413 [2:0]						WSEQ	_ADDR413	[12:0]	-	TA413 [7:0]				FFFFFFFh
R13116 (333Ch)	WSEQ_Sequence_415	WSEQ_D	ATA_WIDTI					START413		WSEQ	_ADDR414	[12:0]						FFFFFFFh
R13118	WSEQ_Sequence_416	WSEQ_D	ATA_WIDTI					START414		WSEQ	_ADDR415	[12:0]		ATA414 [7:0]				FFFFFFFh
(333Eh) R13120	WSEQ_Sequence_417		NSEQ_DEL ATA_WIDTI	AY415 [3:0] H416 [2:0]		WS	EQ_DATA_	START415	[3:0]	WSEQ	_ADDR416	[12:0]		ATA415 [7:0]				FFFFFFFh
(3340h) R13122	WSEQ_Sequence_418		NSEQ_DEL	AY416 [3:0] H417 [2:0]]	WS	EQ_DATA_	START416	[3:0]	WSEQ	_ADDR417	[12:0]	WSEQ_DA	ATA416 [7:0]				FFFFFFFh
(3342h) R13124	WSEQ Sequence 419		NSEQ_DEL ATA_WIDTI	AY417 [3:0 H418 [2:0]		WS	EQ_DATA_	START417	[3:0]	WSEQ	ADDR418	[12:0]	WSEQ_DA	ATA417 [7:0]				FFFFFFFh
(3344h) R13126	WSEQ_Sequence_420	1	_	AY418 [3:0]		WS	EQ_DATA_	START418	[3:0]		ADDR419		WSEQ_DA	ATA418 [7:0]				FFFFFFF
(3346h)			- NSEQ_DEI	AY419 [3:0		WS	EQ_DATA_	START419	[3:0]				WSEQ_DA	ATA419 [7:0]				
R13128 (3348h)	WSEQ_Sequence_421	_	ATA_WIDTI NSEQ_DEI	H420 [2:0] .AY420 [3:0]]	WS	EQ_DATA_	START420	[3:0]	WSEQ	_ADDR420	[12:0]	WSEQ_DA	ATA420 [7:0]				FFFFFFFh
R13130 (334Ah)	WSEQ_Sequence_422	_	ATA_WIDTI NSEQ_DEL	H421 [2:0] .AY421 [3:0]]	WS	EQ_DATA_	START421	[3:0]	WSEQ	_ADDR421	[12:0]	WSEQ_DA	TA421 [7:0]				FFFFFFFh
R13132 (334Ch)	WSEQ_Sequence_423	-	ATA_WIDTI NSEQ DEL	H422 [2:0] AY422 [3:0]	1	WS	EQ DATA	START422	[3:0]	WSEQ	_ADDR422	[12:0]	WSEQ DA	(TA422 [7:0]				FFFFFFFh
R13134 (334Eh)	WSEQ_Sequence_424	WSEQ_D	ATA_WIDTI					START423		WSEQ	_ADDR423	[12:0]	-	ATA423 [7:0]				FFFFFFFh
R13136 (3350h)	WSEQ_Sequence_425	WSEQ_D	ATA_WIDTI	H424 [2:0]						WSEQ	_ADDR424	[12:0]						FFFFFFFh
R13138	WSEQ_Sequence_426		NSEQ_DEL ATA_WIDTI	AY424 [3:0] H425 [2:0]		WS	EQ_DATA_	START424	[3:0]	WSEQ	_ADDR425	[12:0]	WSEQ_DA	ATA424 [7:0]				FFFFFFFh
(3352h) R13140	WSEQ Sequence 427		NSEQ_DEL	AY425 [3:0] H426 [2:0]]	WS	EQ_DATA_	START425	[3:0]	WSEQ	_ADDR426	[12:0]	WSEQ_DA	ATA425 [7:0]				FFFFFFFh
(3354h) R13142	WSEQ Sequence 428		NSEQ_DEL ATA WIDTI	.AY426 [3:0 H427 [2:0]]	WS	EQ_DATA_	START426	[3:0]	WSEQ	ADDR427	[12:0]	WSEQ_DA	ATA426 [7:0]				FFFFFFFh
(3356h)	_ · _		_	AY427 [3:0]	WS	EQ_DATA_	START427	[3:0]		ADDR428		WSEQ_DA	ATA427 [7:0]				
R13144 (3358h)	WSEQ_Sequence_429	- 1	NSEQ_DEL	AY428 [3:0]	WS	EQ_DATA_	START428	[3:0]				WSEQ_DA	TA428 [7:0]				FFFFFFFh
R13146 (335Ah)	WSEQ_Sequence_430	-	ATA_WIDTI NSEQ_DEI	H429 [2:0] _AY429 [3:0]]	WS	EQ_DATA_	START429	[3:0]	WSEQ	_ADDR429	[12:0]	WSEQ_DA	ATA429 [7:0]				FFFFFFFh
R13148 (335Ch)	WSEQ_Sequence_431	-	ATA_WIDTI NSEQ_DEL	H430 [2:0] AY430 [3:0]	1	WS	EQ_DATA	START430	[3:0]	WSEQ	_ADDR430	[12:0]	WSEQ_DA	TA430 [7:0]				FFFFFFFh
R13150 (335Eh)	WSEQ_Sequence_432		ATA_WIDTI NSEQ_DEI	H431 [2:0] .AY431 [3:0]	1	WS	EQ DATA	START431	[3:0]	WSEQ	_ADDR431	[12:0]	WSEQ DA	ATA431 [7:0]				FFFFFFFh
R13152 (3360h)	WSEQ_Sequence_433	WSEQ_D	ATA_WIDTI					START432		WSEQ	_ADDR432	[12:0]		TA432 [7:0]				FFFFFFFh
R13154	WSEQ_Sequence_434	WSEQ_D	ATA_WIDTI	H433 [2:0]						WSEQ	_ADDR433	[12:0]						FFFFFFFh
(3362h) R13156	WSEQ_Sequence_435	WSEQ_D	ATA_WIDTI					START433		WSEQ	_ADDR434	[12:0]		ATA433 [7:0]				FFFFFFFh
(3364h) R13158	WSEQ_Sequence_436		NSEQ_DEL	AY434 [3:0] H435 [2:0]]	WS	EQ_DATA_	START434	[3:0]	WSEQ	_ADDR435	[12:0]	WSEQ_DA	ATA434 [7:0]				FFFFFFFh
(3366h) R13160	WSEQ Sequence 437		NSEQ_DEL ATA WIDTI	AY435 [3:0] H436 [2:0]		WS	EQ_DATA_	START435	[3:0]	WSEQ	ADDR436	[12:0]	WSEQ_DA	ATA435 [7:0]				FFFFFFFh
(3368h)	WSEQ Sequence 438		_	AY436 [3:0		WS	EQ_DATA_	START436	[3:0]		ADDR437		WSEQ_DA	ATA436 [7:0]				FFFFFFF
(336Ah)	_ · _		NSEQ_DEL	AY437 [3:0]	WS	EQ_DATA_	START437	[3:0]		-		WSEQ_DA	ATA437 [7:0]				
R13164 (336Ch)	WSEQ_Sequence_439			AY438 [3:0		WS	EQ_DATA_	START438	[3:0]		_ADDR438		WSEQ_DA	ATA438 [7:0]				FFFFFFFh
R13166 (336Eh)	WSEQ_Sequence_440		ATA_WIDTI NSEQ_DEI	H439 [2:0] .AY439 [3:0]]	WS	EQ_DATA_	START439	[3:0]	WSEQ	_ADDR439	[12:0]	WSEQ_DA	ATA439 [7:0]				FFFFFFFh
R13168 (3370h)	WSEQ_Sequence_441		ATA_WIDTI NSEQ DEL	H440 [2:0] .AY440 [3:0	1	WS	EQ DATA	START440	[3:0]	WSEQ	_ADDR440	[12:0]	WSEQ DA	TA440 [7:0]				FFFFFFFh
R13170 (3372h)	WSEQ_Sequence_442	WSEQ_D	ATA_WIDTI		-			START441		WSEQ	_ADDR441	[12:0]		TA441 [7:0]				FFFFFFFh
R13172	WSEQ_Sequence_443	WSEQ_D	ATA_WIDTI	H442 [2:0]						WSEQ	_ADDR442	[12:0]						FFFFFFFh
	WSEQ_Sequence_444	WSEQ_D	ATA_WIDTI					START442		WSEQ	_ADDR443	[12:0]	_	ATA442 [7:0]				FFFFFFFh
	WSEQ_Sequence_445		NSEQ_DEL	.AY443 [3:0] H444 [2:0]]	WS	EQ_DATA	START443	[3:0]	WSEC	_ADDR444	[12:0]	WSEQ_DA	ATA443 [7:0]				FFFFFFFh
(3378h) R13178	WSEQ Sequence 446		_	AY444 [3:0 H445 [2:0]		WS	EQ_DATA_	START444	[3:0]	WSEG	ADDR445	[12:0]	WSEQ_DA	ATA444 [7:0]				FFFFFFFh
(337Ah) R13180	_ · _	1	_	AY445 [3:0		WS	eq_data_	START445	[3:0]		ADDR446		WSEQ_DA	ATA445 [7:0]				
(337Ch)	WSEQ_Sequence_447	1	NSEQ_DEL	AY446 [3:0]]	WS	EQ_DATA_	START446	[3:0]		-		WSEQ_DA	ATA446 [7:0]				FFFFFFFh
R13182 (337Eh)	WSEQ_Sequence_448		ATA_WIDTI NSEQ_DEL	H447 [2:0] _AY447 [3:0]		WS	EQ_DATA_	START447	[3:0]	WSEQ	_ADDR447	[12:0]	WSEQ_D/	TA447 [7:0]				FFFFFFFh



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	1	7 1	16 0	Default
R13184 (3380h)	WSEQ_Sequence_449	-	ATA_WIDTH NSEQ_DEL	-1448 [2:0] .AY448 [3:0	1	WS	EQ DATA	START448	[3:0]	WSEC	_ADDR448	8 [12:0]	WSEQ D	ATA448 [7:0	1				FFFFFFFh
R13186 (3382h)	WSEQ_Sequence_450	WSEQ_D/	ATA_WIDTI					START449		WSEC	_ADDR449	9 [12:0]	-	ATA449 [7:0					FFFFFFFh
R13188 (3384h)	WSEQ_Sequence_451	WSEQ_D/	ATA_WIDTH							WSEC	_ADDR450) [12:0]			-				FFFFFFFh
R13190	WSEQ_Sequence_452	WSEQ_D/	ATA_WIDTI	-451 [2:0]				START450		WSEC	_ADDR451	[12:0]		ATA450 [7:0					FFFFFFFh
(3386h) R13192	WSEQ_Sequence_453		NSEQ_DEL ATA_WIDTH	.AY451 [3:0 H452 [2:0]]			START451		WSEC	_ADDR452	2 [12:0]	WSEQ_D/	ATA451 [7:0)]				FFFFFFFh
(3388h) R13194	WSEQ_Sequence_454		NSEQ_DEL	.AY452 [3:0 -1453 [2:0]]	WS	EQ_DATA_	START452	[3:0]	WSEC	_ADDR453	8 [12:0]	WSEQ_D/	ATA452 [7:0]				FFFFFFFh
(338Ah) R13196	WSEQ_Sequence_455		NSEQ_DEL ATA_WIDTI	.AY453 [3:0 - 454 [2:0]]	WS	EQ_DATA_	START453	[3:0]	WSEC	ADDR454	[12:0]	WSEQ_D/	ATA453 [7:0)]				FFFFFFFh
(338Ch) R13198	WSEQ Sequence 456	V	_	AY454 [3:0]	WS	EQ_DATA_	START454	[3:0]		ADDR455		WSEQ_D/	ATA454 [7:0)]				FFFFFFF
(338Eh)	_ · _	V	- NSEQ_DEL	AY455 [3:0]	WS	EQ_DATA_	START455	[3:0]				WSEQ_D/	ATA455 [7:0)]				
R13200 (3390h)	WSEQ_Sequence_457	V	_	AY456 [3:0]	WS	EQ_DATA_	START456	[3:0]		_ADDR456		WSEQ_D/	ATA456 [7:0)]				FFFFFFFh
R13202 (3392h)	WSEQ_Sequence_458		ATA_WIDTH NSEQ_DEL	-1457 [2:0] .AY457 [3:0]	WS	EQ_DATA_	START457	[3:0]	WSEC	_ADDR457	7 [12:0]	WSEQ_D/	ATA457 [7:0]				FFFFFFFh
R13204 (3394h)	WSEQ_Sequence_459		ATA_WIDTH NSEQ_DEL	-1458 [2:0] .AY458 [3:0]	WS	EQ_DATA	START458	[3:0]	WSEC	_ADDR458	8 [12:0]	WSEQ_D/	ATA458 [7:0)]				FFFFFFFh
R13206 (3396h)	WSEQ_Sequence_460	_	ATA_WIDTH	-459 [2:0] AY459 [3:0	1			START459		WSEC	_ADDR459	9 [12:0]	WSEQ D	- ATA459 [7:0	1				FFFFFFFh
R13208 (3398h)	WSEQ_Sequence_461	WSEQ_D/	ATA_WIDTI					-		WSEC	_ADDR460) [12:0]		ATA460 [7:0	-				FFFFFFFh
R13210	WSEQ_Sequence_462	WSEQ_D/	ATA_WIDTI	-461 [2:0]				START460		WSEC	_ADDR461	[12:0]		-					FFFFFFFh
(339Ah) R13212	WSEQ_Sequence_463	WSEQ_D/	ATA_WIDTI					START461		WSEC	_ADDR462	2 [12:0]		ATA461 [7:0					FFFFFFFh
(339Ch) R13214	WSEQ_Sequence_464		NSEQ_DEL	.AY462 [3:0 -1463 [2:0]]	WS	EQ_DATA_	START462	[3:0]	WSEC	_ADDR463	8 [12:0]	WSEQ_D/	ATA462 [7:0]				FFFFFFFh
(339Eh) R13216	WSEQ Sequence 465		NSEQ_DEL ATA_WIDTI	.AY463 [3:0 -1464 [2:0]]	WS	EQ_DATA_	START463	[3:0]	WSEC	ADDR464	l [12:0]	WSEQ_D/	ATA463 [7:0]				FFFFFFFh
(33A0h) R13218	WSEQ_Sequence_466	V	_	AY464 [3:0]	WS	EQ_DATA_	START464	[3:0]	WSEC	ADDR465	5[12:0]	WSEQ_D/	ATA464 [7:0]				FFFFFFFh
(33A2h) R13220		V	_	AY465 [3:0]	WS	EQ_DATA_	START465	[3:0]		ADDR466		WSEQ_D/	ATA465 [7:0]				
(33A4h)	WSEQ_Sequence_467	v	- NSEQ_DEL	AY466 [3:0]	WS	EQ_DATA_	START466	[3:0]				WSEQ_D/	ATA466 [7:0]				FFFFFFFh
R13222 (33A6h)	WSEQ_Sequence_468	v	_	AY467 [3:0]	WS	EQ_DATA_	START467	[3:0]		_ADDR467		WSEQ_D/	ATA467 [7:0]				FFFFFFFh
R13224 (33A8h)	WSEQ_Sequence_469	_	ATA_WIDTH NSEQ_DEL	-1468 [2:0] .AY468 [3:0]	WS	EQ_DATA_	START468	[3:0]	WSEC	_ADDR468	8 [12:0]	WSEQ_D/	ATA468 [7:0)]				FFFFFFFh
R13226 (33AAh)	WSEQ_Sequence_470	-	ATA_WIDTH WSEQ_DEL	-1469 [2:0] .AY469 [3:0]	WS	EQ_DATA_	START469	[3:0]	WSEC	_ADDR469	9 [12:0]	WSEQ_D/	ATA469 [7:0]				FFFFFFFh
R13228 (33ACh)	WSEQ_Sequence_471		ATA_WIDTH NSEQ DEL	-1470 [2:0] .AY470 [3:0	1	WS	EQ DATA	START470	[3:0]	WSEC	_ADDR470) [12:0]	WSEQ D	ATA470 [7:0)]				FFFFFFFh
R13230 (33AEh)	WSEQ_Sequence_472		ATA_WIDTH		1	WS		START471	[3:0]	WSEC	_ADDR471	[12:0]	WSEO D	ATA471 [7:0	1				FFFFFFFh
. ,	WSEQ_Sequence_473	WSEQ_D/	ATA_WIDTH	H472 [2:0] AY472 [3:0				START472		WSEC	_ADDR472	2 [12:0]		ATA472 [7:0					FFFFFFFh
R13234 (33B2h)	WSEQ_Sequence_474	WSEQ_D/	ATA_WIDTI	473 [2:0]						WSEC	_ADDR473	8 [12:0]	-						FFFFFFFh
R13236	WSEQ_Sequence_475	WSEQ_D/	ATA_WIDTI					START473		WSEC	_ADDR474	l [12:0]		ATA473 [7:0	-				FFFFFFFh
(33B4h) R13238	WSEQ_Sequence_476		NSEQ_DEL ATA_WIDTH	.AY474 [3:0 -1475 [2:0]]	WS	EQ_DATA_	START474	[3:0]	WSEC	_ADDR475	5 [12:0]	WSEQ_D/	ATA474 [7:0]				FFFFFFFh
(33B6h) R13240	WSEQ Sequence 477		NSEQ_DEL ATA_WIDTI	.AY475 [3:0 -1476 [2:0]]	WS	EQ_DATA_	START475	[3:0]	WSEC	_ADDR476	6 [12:0]	WSEQ_D/	ATA475 [7:0)]				FFFFFFFh
(33B8h) R13242	WSEQ_Sequence_478		NSEQ_DEL	AY476 [3:0]	WS	EQ_DATA_	START476	[3:0]	WSEG	ADDR477	7 [12:0]	WSEQ_D/	ATA476 [7:0)]				FFFFFFFh
(33BAh) R13244	WSEQ Sequence 479	V	_	AY477 [3:0]	WS	EQ_DATA_	START477	[3:0]		ADDR478		WSEQ_D/	ATA477 [7:0]				FFFFFFF
(33BCh)	_ · _	V	NSEQ_DEL	AY478 [3:0]	WS	EQ_DATA_	START478	[3:0]		-		WSEQ_D/	ATA478 [7:0]				
R13246 (33BEh)	WSEQ_Sequence_480	V		AY479 [3:0]	WS	EQ_DATA_	START479	[3:0]		_ADDR479		WSEQ_D/	ATA479 [7:0)]				FFFFFFFh
R13248 (33C0h)	WSEQ_Sequence_481	V	_	AY480 [3:0]	WS	EQ_DATA_	START480	[3:0]		_ADDR480		WSEQ_D	ATA480 [7:0]				FFFFFFFh
R13250 (33C2h)	WSEQ_Sequence_482		ATA_WIDTH NSEQ_DEL	-1481 [2:0] .AY481 [3:0]	WS	EQ_DATA	START481	[3:0]	WSEC	_ADDR481	[12:0]	WSEQ_D	ATA481 [7:0]				FFFFFFFh
R13252 (33C4h)	WSEQ_Sequence_483	WSEQ_D/	ATA_WIDTI					START482		WSEC	_ADDR482	2 [12:0]		ATA482 [7:0					FFFFFFFh
R13254 (33C6h)	WSEQ_Sequence_484	WSEQ_D/	ATA_WIDTI	-1483 [2:0]				-		WSEC	_ADDR483	8 [12:0]			-				FFFFFFFh
(000001)		V	WSEQ_DEL	.AY483 [3:0	J	WS	EQ_DATA_	START483	[3:0]				WSEQ_D/	ATA483 [7:0	IJ				<u> </u>

Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R13256 (33C8h)	WSEQ_Sequence_485	_	DATA_WIDT WSEQ_DE	H484 [2:0] LAY484 [3:0	1	WS	EQ DATA	START484	[3:0]	WSEC	2_ADDR484	1 [12:0]	WSEQ DA	TA484 [7:0]				FFFFFFFh
R13258 (33CAh)	WSEQ_Sequence_486	WSEQ_D	DATA_WIDT				EQ_DATA_			WSEC	_ADDR48	5 [12:0]		TA485 [7:0]				FFFFFFFh
R13260 (33CCh)	WSEQ_Sequence_487	WSEQ_D	DATA_WIDT				EQ DATA	-		WSEC	_ADDR486	6 [12:0]		TA486 [7:0]	-			FFFFFFFh
R13262 (33CEh)	WSEQ_Sequence_488	WSEQ_D	DATA_WIDT				EQ_DATA			WSEC	2_ADDR48	7 [12:0]		TA487 [7:0]				FFFFFFFh
R13264 (33D0h)	WSEQ_Sequence_489	WSEQ_D	DATA_WIDT				EQ_DATA	-		WSEC	2_ADDR488	3 [12:0]	_	TA488 [7:0]				FFFFFFFh
R13266 (33D2h)	WSEQ_Sequence_490	WSEQ_D	DATA_WIDT				EQ_DATA	-		WSEC	2_ADDR48	9 [12:0]		TA489 [7:0]				FFFFFFFh
R13268 (33D4h)	WSEQ_Sequence_491	WSEQ_D	DATA_WIDT				EQ DATA	-		WSEC	2_ADDR490) [12:0]		TA490 [7:0]				FFFFFFFh
R13270 (33D6h)	WSEQ_Sequence_492	WSEQ_D	DATA_WIDT				EQ DATA			WSEC	2_ADDR49	1 [12:0]		TA491 [7:0]				FFFFFFFh
R13272 (33D8h)	WSEQ_Sequence_493	WSEQ_D	DATA_WIDT				EQ DATA			WSEC	2_ADDR492	2 [12:0]		TA492 [7:0]				FFFFFFFh
R13274 (33DAh)	WSEQ_Sequence_494	WSEQ_D	DATA_WIDT				EQ_DATA			WSEC	_ADDR493	3 [12:0]		TA493 [7:0]				FFFFFFFh
R13276 (33DCh)	WSEQ_Sequence_495	WSEQ_D	DATA_WIDT				EQ DATA	-		WSEC	_ADDR494	4 [12:0]		TA494 [7:0]				FFFFFFFh
R13278 (33DEh)	WSEQ_Sequence_496	WSEQ_D	DATA_WIDT				EQ DATA			WSEC	_ADDR49	5 [12:0]		TA494 [7:0]				FFFFFFFh
R13280 (33E0h)	WSEQ_Sequence_497	WSEQ_D	DATA_WIDT	-			EQ_DATA_			WSEC	2_ADDR496	6 [12:0]		TA495 [7:0]				FFFFFFFh
R13282 (33E2h)	WSEQ_Sequence_498	WSEQ_D	DATA_WIDT				EQ_DATA_	-		WSEC	ADDR49	7 [12:0]	_	TA490 [7:0]				FFFFFFFh
R13284 (33E4h)	WSEQ_Sequence_499	WSEQ_D	DATA_WIDT	H498 [2:0]			EQ DATA			WSEC	2_ADDR498	3 [12:0]		TA498 [7:0]				FFFFFFFh
R13286 (33E6h)	WSEQ_Sequence_500	WSEQ_D	DATA_WIDT	LAY498 [3:0 H499 [2:0] LAY499 [3:0						WSEC	2_ADDR499	9 [12:0]		TA498 [7:0]				FFFFFFFh
R13288 (33E8h)	WSEQ_Sequence_501	WSEQ_D	DATA_WIDT	H500 [2:0]			EQ_DATA_			WSEC	ADDR500) [12:0]						FFFFFFFh
R13290 (33EAh)	WSEQ_Sequence_502	WSEQ_D	DATA_WIDT				EQ_DATA_	-		WSEC	2_ADDR50	1 [12:0]		TA500 [7:0]	-			FFFFFFFh
R13292 (33ECh)	WSEQ_Sequence_503	WSEQ_D	DATA_WIDT				EQ_DATA	-		WSEC	_ADDR502	2 [12:0]	WSEQ_DA					FFFFFFFh
R13294 (33EEh)	WSEQ_Sequence_504	WSEQ_D	DATA_WIDT				EQ_DATA	-		WSEC	_ADDR503	3 [12:0]	_	TA502 [7:0]				FFFFFFFh
R13296 (33F0h)	WSEQ_Sequence_505	WSEQ_D	DATA_WIDT				EQ_DATA_			WSEC	_ADDR504	1 [12:0]	WSEQ_DA					FFFFFFFh
R13298 (33F2h)	WSEQ_Sequence_506	WSEQ_D	DATA_WIDT		-		EQ_DATA			WSEC	_ADDR50	5 [12:0]	-	TA504 [7:0]				FFFFFFFh
R13300 (33F4h)	WSEQ_Sequence_507	WSEQ_D	DATA_WIDT				EQ_DATA_			WSEC	ADDR506	6 [12:0]	-	TA505 [7:0]				FFFFFFFh
R13302	WSEQ_Sequence_508	WSEQ_D	DATA_WIDT				EQ_DATA_			WSEC	_ADDR507	7 [12:0]	-	TA506 [7:0]				FFFFFFFh
(33F6h) R131076 (20004h)	OTP_HPDET_Cal_1		WSEQ_DE	LAY507 [3:0	HP_OFFS	ET_11 [7:0]		STAR1507	[3:0]				HP_OFFS	TA507 [7:0] ET_10 [7:0]	-			00000000h
. ,	OTP_HPDET_Cal_2					= 2 [7:0]							SPARI	ET_00 [7:0] E1 [7:0]				00000000h
R294912	EVENTLOG1_ CONTROL	0	0	0	IP_GRADIE 0	0	0	0	0	0	0	0	HP_GRADIE	0	0	0	0	00000000h
. ,	EVENTLOG1_TIMER_	0	0	0	0	0	0	0	0	0	0	0	0	0	0		0 EVENTLO G1_ENA 0	00000000h
(48004h)	SEL	0	0	0	0	0	0	0	0	0	0	0	0	0	0	EVEN	TLOG1_ _SEL [1:0]	000000000
R294924 (4800Ch)	EVENTLOG1_FIFO_ CONTROL1	0	0	0	0	0	0	0	0	0	0	0	0	0 EVEN	0 NTLOG1_FI	0	0	00000001h
R294926 (4800Eh)	EVENTLOG1_FIFO_ POINTER1	0	0	0	0	0	0	0	0	0	0	0	0	0	_		EVENTLO G1 NOT	0000000h
		0	0	0	0	EVE	NTLOG1 F	FIFO WPTF	R [3:0]	0	0	0	0	EVE	NTLOG1 F	STS		
R294944 (48020h)	EVENTLOG1_CH_ ENABLE1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0 EVENTLO G1_CH1_	00000000h
D004040		ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	ENA	
R294948 (48024h)	EVENTLOG1_EVENT_ STATUS	0 EVENTLO G1 CH16	0 EVENTLO G1 CH15	0 EVENTLO G1 CH14	0 EVENTLO G1 CH13	0 EVENTLO G1 CH12	0 EVENTLO G1 CH11	0 EVENTLO G1 CH10	0 EVENTLO G1 CH9	0 EVENTLO G1 CH8	0 EVENTLO G1 CH7	0 EVENTLO G1 CH6	0 EVENTLO G1 CH5	0 EVENTLO G1 CH4	0 EVENTLO G1 CH3	0 EVENTLC G1 CH2	0 DEVENTLO G1_CH1_	00000000h
		STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	STS	



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	10	
R294976 (48040h)	EVENTLOG1_CH1_ DEFINE	G1 CH1	0 EVENTLO G1_CH1_	G1 CH1	0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH1_SEL	0 [9:0]	0	0	0	00000000h
R294978 (48042h)	EVENTLOG1_CH2_ DEFINE	DB 0 EVENTLC G1 CH2	POL 0 EVENTLO G1_CH2_	FILT - 0 EVENTLO G1 CH2	0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH2_SEL	0 [9:0]	0	0	0	00000000h
	EVENTLOG1_CH3_ DEFINE	DB 0 EVENTLC	POL 0 EVENTLO	FILT 0 EVENTLO	0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH3_SEL	0 [9:0]	0	0	0	00000000h
R294982 (48046h)	EVENTLOG1_CH4_ DEFINE	DB 0	POL 0	G1_CH3_ FILT 0	0	0	0	0	0	0	0	0	0	0	0	0	0	00000000h
. ,	EVENTLOG1 CH5	G1_CH4_ DB 0	EVENTLO G1_CH4_ POL 0	EVENTLO G1_CH4_ FILT 0	0	0	0	0	0	0	EV	ENILOG1_	CH4_SEL	0	0	0	0	00000000h
(48048h)	DEFINE	G1_CH5_ DB	POL	G1_CH5_ FILT	0	0	0	-			EV		CH5_SEL					
R294986 (4804Ah)	EVENTLOG1_CH6_ DEFINE	0 EVENTLC G1_CH6_ DB	0 EVENTLO G1_CH6_ POL	0 EVENTLO G1_CH6_ FILT	0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH6_SEL	0 [9:0]	0	0	0	00000000h
	EVENTLOG1_CH7_ DEFINE	0 EVENTLC G1_CH7_ DB	0 EVENTLO G1_CH7_ POL	0 EVENTLO G1_CH7_ FILT	0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH7_SEL	0 [9:0]	0	0	0	00000000h
	EVENTLOG1_CH8_ DEFINE	0 EVENTLC	0 EVENTLO	0	0	0 0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH8_SEL	0 [9:0]	0	0	0	00000000h
R294992 (48050h)	EVENTLOG1_CH9_ DEFINE	0 EVENTLC	0 EVENTLO G1_CH9_ POL	0 EVENTLO	0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 CH9_SEL	0 [9:0]	0	0	0	00000000h
	EVENTLOG1_CH10_ DEFINE	0	0 EVENTLO G1_CH10_ POL	0	0	0	0	0	0	0	0 EVE	0 ENTLOG1_	0 CH10_SEL	0 [9:0]	0	0	0	00000000h
R294996 (48054h)	EVENTLOG1_CH11_ DEFINE	0 EVENTLC	0 EVENTLO G1_CH11_ POL	0 EVENTLO	0	0	0	0	0	0	0 EVE	0 ENTLOG1_	0 CH11_SEL	0 [9:0]	0	0	0	00000000h
	EVENTLOG1_CH12_ DEFINE	0	0 EVENTLO G1_CH12 POL	0	0	0	0	0	0	0	0 EVE	0 ENTLOG1_	0 CH12_SEL	0 [9:0]	0	0	0	00000000h
	EVENTLOG1_CH13_ DEFINE	0	0 EVENTLO G1_CH13_ POL	0	0	0	0	0	0	0	0 EVE	0 ENTLOG1_	0 CH13_SEL	0 [9:0]	0	0	0	00000000h
R295002 (4805Ah)	EVENTLOG1_CH14_ DEFINE	0	0 EVENTLO G1_CH14_ POL	0	0	0	0	0	0	0	0 EVE	0 ENTLOG1_	0 CH14_SEL	0 [9:0]	0	0	0	00000000h
R295004 (4805Ch)	EVENTLOG1_CH15_ DEFINE	0	0 EVENTLO G1_CH15_ POL	0	0	0	0	0	0	0	0 EVE	0 ENTLOG1_	0 CH15_SEL	0 [9:0]	0	0	0	00000000h
R295006 (4805Eh)	EVENTLOG1_CH16_ DEFINE	0 EVENTLC	0 EVENTLO G1_CH16_ POL	0 EVENTLO	0	0	0	0	0	0	0 EVE	0 NTLOG1_	0 CH16_SEL	0 [9:0]	0	0	0	00000000h
R295040 (48080h)	EVENTLOG1_FIFO0_ READ	0	0	0	0 EVENTLO G1_ FIFO0	0	0	0	0	0	0 EV	0 ENTLOG1_	0 _FIFO0_ID	0 [9:0]	0	0	0	00000000h
R295042 (48082h)	EVENTLOG1_FIFO0_ TIME				POL ⁻					FO0_TIME								00000000h
R295044 (48084h)	EVENTLOG1_FIFO1_ READ	0	0	0	0 EVENTLO G1_ FIFO1_	0	0	0	0	0	0	0 ENTLOG1_	0 _FIFO1_ID [0 [9:0]	0	0	0	00000000h
	EVENTLOG1_FIFO1_ TIME		I	1	POL ⁻		1			FO1_TIME								00000000h
R295048 (48088h)	EVENTLOG1_FIFO2_ READ	0	0	0	0 EVENTLO G1_ FIFO2_	0	0	0	0	0	0	0 ENTLOG1_	0 _FIFO2_ID	0 [9:0]	0	0	0	00000000h
R295050 (4808Ah)	EVENTLOG1_FIFO2_				POL					FO2_TIME								00000000h



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	1	7 1	16 0	Default
R295052 (4808Ch)	EVENTLOG1_FIFO3_ READ	0	0	0	0 EVENTLO	0	0	0	0	0	0 EV	0 ENTLOG1		0 9-01	0	0)	0	00000000h
(,		0	0	Ū	G1_ FIFO3_ POL	Ū	0				Ľv	LINILOO1_	1100_101	5.0]					
R295054 (4808Eh)	EVENTLOG1_FIFO3_ TIME									FO3_TIME									00000000h
R295056 (48090h)	EVENTLOG1_FIFO4_	0	0	0	0 EVENTLO	0	0	0	0	0	0	0 ENTLOG1		0	0	0)	0	00000000h
(4000011)		0	U	U	G1_ FIFO4_ POL	U	U					ENTLOGI_	FIF04_ID	9.0]					
	EVENTLOG1_FIFO4_ TIME									FO4_TIME									00000000h
R295060 (48094h)	EVENTLOG1_FIFO5_	0	0	0	0 EVENTLO	0	0	0	0	0	0	0 ENTLOG1		0	0	0)	0	00000000h
. ,		0	0	0	G1_ FIF05_ POL	U	0					ENTLOGI_	FIF05_ID	9.0]					
R295062 (48096h)	EVENTLOG1_FIFO5_ TIME									FO5_TIME									00000000h
R295064 (48098h)	EVENTLOG1_FIFO6_ RFAD	0	0	0	0 EVENTLO	0	0	0	0	0	0	0 ENTLOG1		0	0	0)	0	00000000h
. ,		0	0	U	G1_ FIFO6_ POL	U	0					ENTLOGI_	_FIFO0_ID	9.0]					
	EVENTLOG1_FIFO6_ TIME								_	FO6_TIME									00000000h
R295068 (4809Ch)	EVENTLOG1_FIF07_	0	0	0		0	0	0	0	0	0	0	0	0	0	0)	0	0000000h
(4009011)	READ	0	0	0	EVENTLO G1_ FIFO7_ POL	0	0				EV	'ENTLOG1_	FIFO7_ID	9:0]					
	EVENTLOG1_FIFO7_ TIME									F07_TIME									00000000h
R295072	EVENTLOG1_FIF08_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0)	0	0000000h
(480A0h)	READ	0	0	0	EVENTLO G1_ FIFO8_ POL	0	0				EV	'ENTLOG1_	FIFO8_ID	9:0]					
	EVENTLOG1_FIFO8_ TIME									FO8_TIME									00000000h
R295076	EVENTLOG1_FIFO9_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0)	0	00000000h
(480A4h)	READ	0	0	0	EVENTLO G1_ FIFO9_ POL	0	0				EV	'ENTLOG1_	FIFO9_ID	9:0]					
	EVENTLOG1_FIFO9_ TIME			•						FO9_TIME									00000000h
R295080	EVENTLOG1_FIFO10_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0)	0	0000000h
(480A8h)	READ	0	0	0	EVENTLO G1_ FIFO10_ POL	0	0				EV	ENTLOG1_	FIFO10_ID	[9:0]					
R295082 (480AAh)	EVENTLOG1_FIFO10_ TIME									FO10_TIME FO10_TIME									00000000h
R295084 (480ACh)	EVENTLOG1_EIE011	0	0	0		0	0	0	0	0	0	0	0	0	0	0)	0	0000000h
		0	0	0	EVENTLO G1 FIFO11_ POL	0	0					ENTLOG1_	FIF011_ID	[9:0]					
R295086 (480AEh)	EVENTLOG1_FIFO11_ TIME									FO11_TIME FO11_TIME									00000000h
· ,	EVENTLOG1_EIE012	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0)	0	0000000h
(4000011)	READ	0	0	0	EVENTLO G1_ FIFO12_ POL	0	0				EV	ENTLOG1_	FIFO12_ID	[9:0]					
R295090 (480B2h)	EVENTLOG1_FIFO12_ TIME									FO12_TIME FO12_TIME	<u> </u>								00000000h
R295092	EVENTLOG1_FIFO13_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0)	0	0000000h
(480B4h)	READ	0	0	0	EVENTLO G1_ FIFO13_ POL	0	0				EVI	ENTLOG1_	FIFO13_ID	[9:0]					
R295094 (480B6h)	EVENTLOG1_FIFO13_ TIMF			•			•			FO13_TIME FO13_TIME									0000000h
R295096	EVENTLOG1_EIE014	0	0	0	0	0	0	0	0	0 0	0	0	0	0	0	0)	0	00000000h
(480B8h)	READ	0	0	0	EVENTLO G1_ FIFO14_ POL	0	0				EV	ENTLOG1_	FIFO14_ID	[9:0]			.		
R295098	EVENTLOG1_FIFO14_			1			1			O14_TIME									0000000h
(480BAh)								EVEN	NILOG1_FI	FO14_TIME	[15:0]								



Table 6-2.	Register Mag	Definition—32-bit regi	on (Cont.)
	rtogiotor map		

Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
	EVENTLOG1_FIFO15_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	00000000h
(480BCh)	READ	0	0	0	EVENTLO G1_ FIFO15_ POL	0	0				EV	ENTLOG1_	FIFO15_ID	[9:0]				
R295102 (480BEh)	EVENTLOG1_FIFO15_ TIME			1					TLOG1_FIF									00000000h
R303104	ALM1_CFG	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	00000000h
(4A000h)		0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	ALM1_ TIMER_ SEL	
R303120 (4A010h)	ALM1_CONFIG1	0	0	0	0	0	0	0	0	0	0	0	0 ALM1_ CH1_ CONT	0	0	0 ALM1_C MOE	0 H1_TRIG_ DE[1:0]	00000000h
	ALM1_CTRL1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000000h
(4A012h)		ALM1_ CH1_UPD	0	0	0	0	0	0	0	0	0	0	ALM1_ CH1_ STOP	0	0	0	ALM1_ CH1_ START	
R303124 (4A0124h)	ALM1_TRIG_VAL1								M1_CH1_TF .M1 CH1 T									00000000h
R303126 (4A016h)	ALM1_PULSE_DUR1								1_CH1_PU 11 CH1 PU		-							00000000h
R303128	ALM1_STATUS1	0	0	0	0	0	0	0	0	0	0.0]	0	0	0	0	0	0	00000000h
(4A018h)		0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	ALM1_ CH1_STS	
R303136 (4A020h)	ALM1_CONFIG2	0	0	0	0	0	0	0	0	0	0	0	0 ALM1_ CH2_	0	0	0 ALM1_C MOE	0 H2_TRIG_ DE[1:0]	00000000h
R303138	ALM1_CTRL2	0	0	0	0	0	0	0	0	0	0	0	CONT 0	0	0	0	0	00000000h
(4A022h)		ALM1_ CH2_UPD	0	0	0	0	0	0	0	0	0	0	ALM1_ CH2_ STOP	0	0	0	ALM1_ CH2_ START	
R303140 (4A024h)	ALM1_TRIG_VAL2								M1_CH2_TF .M1 CH2 T		•							00000000h
R303142 (4A026h)	ALM1_PULSE_DUR2							ALN	1_CH2_PU 1_CH2_PU 11_CH2_PU	LSE_DUR[31:16]							00000000h
R303144 (4A028h)	ALM1_STATUS2	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0 ALM1 CH2 STS	00000000h
R303152 (4A030h)	ALM1_CONFIG3	0	0	0	0	0	0 0	0	0	0 0	0	0	0 ALM1_ CH3	0	0	0 ALM1_C MOE	0 H3_TRIG_ DE[1:0]	00000000h
R303154	ALM1 CTRL3	0	0	0	0	0	0	0	0	0	0	0	CONT 0	0	0	0	0	00000000h
(4A032h)	ALMI_CIRES	ALM1 CH3_UPD	0	0	0	0	0	0	0	0	0	0	ALM1_ CH3_ STOP	0	0	0	ALM1_ CH3_ START	000000000
R303156 (4A034h)	ALM1_TRIG_VAL3			L			I		M1_CH3_TF .M1 CH3 T			1			1	L	1	00000000h
R303158	ALM1_PULSE_DUR3							ALN	1_CH3_PU	LSE_DUR[31:16]							00000000h
(4A036h) R303160	ALM1_STATUS3	0	0	0	0	0	0	ALN 0	11_CH3_PU 0	ILSE_DUR[0	15:0]	0	0	0	0	0	0	00000000h
(4A038h)		0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	ALM1 CH3 STS	
R303168 (4A040h)	ALM1_CONFIG4	0	0	0	0	0	0	0	0	0	0	0	0 ALM1_ CH4	0	0	0 ALM1_C	0 H4_TRIG_ DE[1:0]	00000000h
D000470		0	0	0	0	0	0	0	0	0	0	0	CONT	0	0			00000000
R303170 (4A042h)	ALM1_CTRL4	0 ALM1 CH4_UPD	0	0	0	0	0	0	0	0	0	0	0 ALM1_ CH4_ STOP	0	0	0	0 ALM1_ CH4_ START	00000000h
R303172 (4A044h)	ALM1_TRIG_VAL4			L			I		M1_CH4_TF			1			1	L	I	00000000h
R303174 (4A046h)	ALM1_PULSE_DUR4							ALN	.M1_CH4_T 1_CH4_PU 11_CH4_PU	LSE_DUR[31:16]							00000000h
R303176	ALM1_STATUS4	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	00000000h
(4A048h) R311296	Timer1_Control	0	0	0	0	0	0	0	0	0	0		0 TIMER1_	0	0 TIMER	0 R1_PRESCA	ALM1 CH4_STS ALE [2:0]	00000000h
(4C000h)		0	TIMER1	_REFCLK_	DIV [2:0]	0	TIMER1_	REFCLK_F	REQ_SEL	0	0	CONTINŪ OUS 0	DIR ⁻	ті	MER1_REF	CLK_SRC	[3:0]	
R311298	Timer1_Count_Preset					I	I	[2:0] TIN	IER1_MAX_	COUNT [3	1:16]	I	I	I				00000000h
(4C002h)	Timor 1 Start and Star	0	0	0	0	0	0	TII 0	MER1_MAX	_COUNT [1	5:0]	0	0	0	0	0	0	0000000-
R311302 (4C006h)	Timer1_Start_and_Stop	0	0	0	0	0	0	0	0	0	0	0	U TIMER1_ STOP	0	0	0	U TIMER1_ START	00000000h



Register	Name	31	30	29	28	27	26 10	25	24 8	23 7	22	21	20	19	18	17	16 0	Default
R311304	Timer1 Status	15 0	14 0	13 0	12 0	11 0	0	9 0	0	0	6 0	5 0	4 0	3 0	2 0	1	0	00000000h
(4C008h)	_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	TIMER1_ RUNNING _STS	
R311306 (4C00Ah)	Timer1_Count_ Readback								IER1_CUR_ /IER1_CUR									00000000h
R311308 (4C00Ch)	Timer1_DSP_Clock_ Config	0	0	0	0	0	0	0 TIMER	0 1_DSPCLK	0 _FREQ_SE	0 L [15:0]	0	0	0	0	0	0	00000000h
R311310 (4C00Eh)	Timer1_DSP_Clock_ Status	0	0	0	0	0	0	0 TIMER	0 1_DSPCLK	0 _FREQ_ST	0 S [15:0]	0	0	0	0	0	0	00000000h
R315392 (4D000h)	DSPGP_Status_1	0 DSPGP16			0 DSPGP13	0 DSPGP12		0 DSPGP10		0 DSPGP8_	0 DSPGP7_	0 DSPGP6_	0 DSPGP5_	0 DSPGP4_		0 DSPGP2_	0 DSPGP1_	00000000h
R315424	DSPGP_SET1_Mask_1	_STS 0	_STS 0	_STS 0	_STS 0	_STS 0	_STS 0	_STS 0	STS 0	STS 0	STS 0	STS 0	STS 0	STS 0	STS 0	STS 0	STS 0	0000FFFFh
(4D020h)		DSPGP16 _SET1_ _MASK	_SET1_ MASK	_SET1_ MASK	_SET1_ MASK	DSPGP12 _SET1_ _MASK	_SET1_ MASK	DSPGP10 _SET1_ _MASK	SET1_ MASK	DSPGP8_ SET1_ MASK	DSPGP7_ SET1_ MASK	DSPGP6_ SET1_ MASK	DSPGP5_ SET1_ MASK	DSPGP4_ SET1_ MASK	DSPGP3_ SET1_ MASK	DSPGP2_ SET1_ MASK	DSPGP1_ SET1_ MASK	
R315432 (4D028h)	DSPGP_SET1_ Direction_1	0 DSPGP16			0 DSPGP13				0 DSPGP9_	0 DSPGP8_	0 DSPGP7_	0 DSPGP6_	0 DSPGP5_	0 DSPGP4_	0 DSPGP3_	0 DSPGP2_	0 DSPGP1_	0000FFFFh
R315440	DSPGP SET1 Level 1	_SET1_ DIR 0	_SET1_ DIR 0	_SET1_ DIR_ 0	_SET1_ DIR 0	_SET1_ DIR 0	_SET1_ DIR 0	_SET1_ DIR 0	SEI1_DIR	SET1_DIR 0	0	SET1_DIR	0	SET1_DIR		SETT_DIR 0	SET1_DIR	00000000h
(4D030h)		DSPGP16 SET1	DSPGP15 _SET1_	DSPGP14 _SET1_	DSPGP13 SET1	DSPGP12 _SET1_	DSPGP11 SET1	DSPGP10 SET1	DSPGP9	DSPGP8	DSPGP7	DSPGP6	DSPGP5	DSPGP4 SET1_LVL	DSPGP3	DSPGP2	DSPGP1	00000000
R315456	DSPGP_SET2_Mask_1	- LVL - 0	- LVL -	_ LVL0	- LVL - 0	_ LVL0	- LVL - 0	- LVL - 0	0	0	0	0	0	0	0	0	0	0000FFFFh
(4D040h)		DSPGP16 _SET2_ _MASK	DSPGP15 _SET2_ MASK	DSPGP14 _SET2_ MASK	DSPGP13 _SET2_ MASK	DSPGP12 _SET2_ _MASK	DSPGP11 _SET2_ _MASK	DSPGP10 _SET2_ _MASK	DSPGP9_ SET2_ MASK	DSPGP8_ SET2_ MASK	DSPGP7_ SET2_ MASK	DSPGP6_ SET2_ MASK	DSPGP5_ SET2_ MASK	DSPGP4_ SET2_ MASK	DSPGP3_ SET2_ MASK	DSPGP2_ SET2_ MASK	DSPGP1_ SET2_ MASK	
R315464 (4D048h)	DSPGP_SET2_ Direction 1	0	0	0	0	0 DSPGP12	0	0 DSPGP10	0	0 DSPGP8	0 DSPGP7	0 DSPGP6	0 DSPGP5	0 DSPGP4	0 DSPGP3	0	0 DSPGP1	0000FFFFh
R315472	DSPGP SET2 Level 1	DSPGP16 _SET2_ DIR 0	_SET2_ DIR	_SET2_ DIR 0	_SET2_ DIR 0	_SET2_ DIR	_SET2_ DIR 0	_SET2_ DIR	SET2_DIR	SET2_DIR	SET2_DIR	SET2_DIR	DSPGP5 SET2_DIR	SET2_DIR			SET2_DIR	00000000h
(4D050h)		DSPGP16 _SET2_ _LVL	-	-		DSPGP12 _SET2_ _LVL		DSPGP10 _SET2_ _LVL	DSPGP9	DSPGP8	DSPGP7	DSPGP6	DSPGP5	DSPGP4 SET2_LVL	DSPGP3	DSPGP2	DSPGP1	00000000
R315488 (4D060h)	DSPGP_SET3_Mask_1	0 DSPGP16	0	0 DSPGP14	0 DSPGP13	0 DSPGP12	0	0 DSPGP10	0 DSPGP9_	0 DSPGP8	0 DSPGP7	0 DSPGP6	0 DSPGP5	0 DSPGP4	0 DSPGP3	0 DSPGP2	0 DSPGP1	0000FFFFh
(_SET3_ MASK	_SET3_ MASK	SET3 MASK	SET3_ MASK	_SET3_ MASK	_SET3_ MASK	_SET3_ MASK	SET3 MASK									
R315496 (4D068h)	DSPGP_SET3_ Direction_1	0 DSPGP16		0 DSPGP14	0 DSPGP13	0 DSPGP12		0 DSPGP10	0 DSPGP9_	0 DSPGP8_	0 DSPGP7_	0 DSPGP6_	0 DSPGP5_	0 DSPGP4_	0 DSPGP3_	0 DSPGP2_	0 DSPGP1_	0000FFFFh
R315504	DSPGP SET3 Level 1	_SET3_ DIR 0	_SET3_ DIR 0	_SET3_ DIR 0	_SET3_ DIR 0	_SET3_ DIR 0	_SET3_ DIR 0	_SET3_ DIR 0	SET3_DIR	SET3_DIR 0	SET3_DIR	SET3_DIR 0	0	SET3_DIR	SET3_DIR 0	SET3_DIR 0	SET3_DIR	00000000h
(4D070h)	DSPGP_SET3_Level_T	DSPGP16 _SET3_ _LVL		DSPGP14 _SET3_ _LVL	DSPGP13 _SET3_ _LVL	DSPGP12 _SET3_ _LVL		DSPGP10 _SET3_ _LVL	DSPGP9	DSPGP8 SET3_LVL	DSPGP7	DSPGP6	DSPGP5	DSPGP4 SET3_LVL	DSPGP3	DSPGP2	DSPGP1	00000000
R315520 (4D080h)	DSPGP_SET4_Mask_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000FFFFh
. ,		DSPGP16 _SET4_ _MASK	_SET4_ MASK	DSPGP14 _SET4_ MASK	DSPGP13 _SET4_ MASK	DSPGP12 _SET4_ _MASK	DSPGP11 _SET4_ _MASK	DSPGP10 _SET4_ _MASK	DSPGP9_ SET4_ MASK	DSPGP8_ SET4_ MASK	DSPGP7_ SET4_ MASK	DSPGP6_ SET4_ MASK	DSPGP5_ SET4_ MASK	DSPGP4_ SET4_ MASK	DSPGP3_ SET4_ MASK	DSPGP2_ SET4_ MASK	DSPGP1_ SET4_ MASK	
R315528 (4D088h)	DSPGP_SET4_ Direction_1	0 DSPGP16 SET4	0 DSPGP15 SET4	0 DSPGP14 SET4	0 DSPGP13 SET4	0 DSPGP12 SET4	0 DSPGP11 SET4	0 DSPGP10 SET4	0 DSPGP9 SET4 DIR	0 DSPGP8 SET4 DIR	0 DSPGP7 SET4 DIR	0 DSPGP6 SET4 DIR	0 DSPGP5 SET4 DIR	0 DSPGP4 SET4_DIR	0 DSPGP3 SET4 DIR	0 DSPGP2 SET4 DIR	0 DSPGP1 SET4 DIR	0000FFFFh
R315536	DSPGP SET4 Level 1	 	-DIR- 0	DIR0	DIR 0		DIR 0		0	0	0	0	0	0	0	0	0	00000000h
(4D090h)		DSPGP16 _SET4_ _LVL	_SET4_	DSPGP14 _SET4_ _LVL	DSPGP13 _SET4_ _LVL	DSPGP12 _SET4_	DSPGP11 _SET4_ _LVL	DSPGP10 _SET4_ _LVL	DSPGP9 SET4_LVL	DSPGP8 SET4_LVL	DSPGP7_ SET4_LVL	DSPGP6 SET4_LVL	DSPGP5 SET4_LVL	DSPGP4 SET4_LVL	DSPGP3 SET4_LVL	DSPGP2 SET4_LVL	DSPGP1_ SET4_LVL	
R315552 (4D0A0h)	DSPGP_SET5_Mask_1	0	UVL 0	0	0	UVL 0	0	0	0	0	0	0	0	0	0	0	0	0000FFFFh
(400A011)		DSPGP16 _SET5_ _MASK	DSPGP15 _SET5_ _MASK	DSPGP14 _SET5_ _MASK	DSPGP13 _SET5_ MASK	DSPGP12 _SET5_ _MASK	DSPGP11 _SET5_ _MASK	DSPGP10 _SET5_ _MASK	DSPGP9_ SET5_ MASK	DSPGP8_ SET5_ MASK	DSPGP7_ SET5_ MASK	DSPGP6_ SET5_ MASK	DSPGP5_ SET5_ MASK	DSPGP4_ SET5_ MASK	DSPGP3_ SET5_ MASK	DSPGP2_ SET5_ MASK	DSPGP1_ SET5_ MASK	
R315560 (4D0A8h)	DSPGP_SET5_ Direction_1	0 DSPGP16 _SET5_	0 DSPGP15 _SET5_	0 DSPGP14 _SET5_	0 DSPGP13 _SET5_	0 DSPGP12 _SET5_	0 DSPGP11 SET5	0 DSPGP10 SET5	0 DSPGP9 SET5 DIR	0 DSPGP8 SET5 DIR	0 DSPGP7 SET5 DIR	0 DSPGP6 SET5 DIR	0 DSPGP5 SET5 DIR	0 DSPGP4 SET5 DIR	0 DSPGP3 SET5 DIR	0 DSPGP2 SET5 DIR	0 DSPGP1 SET5 DIR	0000FFFFh
R315568	DSPGP SET5 Level 1	DIR 0	_ DIR0	_ DIR0	DIR [®]	DIR 0	_ DIR0	_ DIR0	0	0	0	0	0	0	0	0	0	00000000h
(4D0B0h)		DSPGP16 _SET5_ _LVL	DSPGP15 _SET5_ _LVL	DSPGP14 _SET5_ _LVL	DSPGP13 _SET5_ _LVL	DSPGP12 _SET5_ _LVL	DSPGP11 _SET5_ _LVL	DSPGP10 _SET5_ _LVL	DSPGP9 SET5_LVL	DSPGP8 SET5_LVL	DSPGP7_ SET5_LVL	DSPGP6 SET5_LVL	DSPGP5_ SET5_LVL	DSPGP4 SET5_LVL	DSPGP3 SET5_LVL	DSPGP2 SET5_LVL	DSPGP1_ SET5_LVL	
R315584 (4D0C0h)	DSPGP_SET6_Mask_1	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000FFFFh
(120001)		DSPGP16 _SET6_ _MASK	DSPGP15 _SET6_ _MASK	DSPGP14 _SET6_ MASK	DSPGP13 _SET6_ MASK	DSPGP12 _SET6_ _MASK	DSPGP11 SET6 _MASK	DSPGP10 _SET6_ _MASK	DSPGP9_ SET6_ MASK	DSPGP8_ SET6_ MASK	DSPGP7_ SET6_ MASK	DSPGP6_ SET6_ MASK	DSPGP5_ SET6_ MASK	DSPGP4_ SET6_ MASK	DSPGP3_ SET6_ MASK	DSPGP2_ SET6_ MASK	DSPGP1_ SET6_ MASK	
	DSPGP_SET6_ Direction_1	0 DSPGP16		0 DSPGP14	0 DSPGP13	0 DSPGP12			0 DSPGP9_	0 DSPGP8_	0 DSPGP7_	0 DSPGP6_	0 DSPGP5	0 DSPGP4_	0 DSPGP3_	0 DSPGP2_	0 DSPGP1_	0000FFFFh
Deter		_SET6_ DIR	_SET6_ DIR	_SET6_ DIR	_SET6_ DIR	_SET6_ DIR	_SET6_ DIR	_SET6_ DIR	SET6_DIR									
R315600 (4D0D0h)	DSPGP_SET6_Level_1	0 DSPGP16 _SET6_	SET6	SET6	SET6	0 DSPGP12 _SET6_	SET6	_SET6_	0 DSPGP9 SET6_LVL	0 DSPGP8_ SET6_LVL	0 DSPGP7_ SET6_LVL	0 DSPGP6_ SET6_LVL	0 DSPGP5_ SET6_LVL	0 DSPGP4 SET6_LVL	0 DSPGP3 SET6_LVL	0 DSPGP2_ SET6_LVL	0 DSPGP1_ SET6_LVL	00000000h
		LVL	LVL	LVL	LVL	LVL	LVL	-LVL		_		_						



00000000h

CS42L92 6 Register Map

				able 6		-	-				-	-	-				1 10	
Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R315616 (4D0E0h)	DSPGP_SET7_Mask_1	0 DSPGP16 _SET7_	0 DSPGP15 _SET7	0 DSPGP14 _SET7_	0 DSPGP13 	0 DSPGP12 	0 DSPGP11 	0 DSPGP10 	0 DSPGP9_ SET7_	0 DSPGP8_ SET7_	0 DSPGP7_ SET7_	0 DSPGP6_ SET7_	0 DSPGP5_ SET7_	0 DSPGP4_ SET7_	0 DSPGP3_ SET7_	0 DSPGP2_ SET7_ MASK	0 DSPGP1_ SET7 MASK	0000FFFFh
	DSPGP_SET7_	MASK 0	MASK 0	MASK 0	MASK 0	MASK 0	MASK 0	MASK 0	0	0	0000FFFF							
4D0E8h)	Direction_1	DSPGP16 _SET7_ DIR	DSPGP15 _SET7_ DIR	DSPGP14 _SET7_ DIR	DSPGP13 _SET7_ _DIR	DSPGP12 _SET7_ _DIR	DSPGP11 _SET7_ DIR	DSPGP10 _SET7_ _DIR	DSPGP9 SET7_DIR	DSPGP8 SET7_DIR	DSPGP7 SET7_DIR	DSPGP6 SET7_DIR	DSPGP5 SET7_DIR	DSPGP4 SET7_DIR	DSPGP3 SET7_DIR	DSPGP2 SET7_DIF	DSPGP1 RSET7_DIR	
R315632 4D0F0h)	DSPGP_SET7_Level_1	0 DSPGP16 SET7	0 DSPGP15 SET7	0 DSPGP14 SET7	0 DSPGP13 SET7	0 DSPGP12 SET7	0 DSPGP11 SET7	0 DSPGP10 _SET7_	0 DSPGP9 SET7 11/1	0 DSPGP8 SET7 11/1	0 DSPGP7 SET7 11/1	0 DSPGP6 SET7 1/4	0 DSPGP5 SET7 11/1	0 DSPGP4 SET7 11/1	0 DSPGP3	0 DSPGP2	0 DSPGP1 SET7 LVL	00000001
R315648	DSPGP SET8 Mask 1						-LVL -	-LVL -	0	0	0	0	0	0	0	0	0	0000FFFFI
4D100h)		DSPGP16 _SET8_ _MASK	DSPGP15 _SET8_ _MASK	DSPGP14 _SET8_ _MASK	DSPGP13 _SET8_ _MASK	DSPGP12 _SET8_ _MASK	DSPGP11 _SET8_ _MASK	DSPGP10 _SET8_ _MASK	DSPGP9_ SET8_ MASK	DSPGP8_ SET8_ MASK	DSPGP7_ SET8_ MASK	DSPGP6_ SET8_ MASK	DSPGP5_ SET8_ MASK	DSPGP4_ SET8_ MASK	DSPGP3_ SET8_ MASK	DSPGP2 SET8 MASK	DSPGP1_ SET8_ MASK	00001111
	DSPGP_SET8_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0000FFFFI
4D108h)	Direction_1	DSPGP16 _SET8_ _DIR	DSPGP15 _SET8_ _DIR	DSPGP14 _SET8_ _DIR	DSPGP13 _SET8_ _DIR	DSPGP12 _SET8_ _DIR	DSPGP11 _SET8_ _DIR	DSPGP10 _SET8_ _DIR	DSPGP9 SET8_DIR	DSPGP8_ SET8_DIR	DSPGP7 SET8_DIR	DSPGP6 SET8_DIR	DSPGP5 SET8_DIR	DSPGP4 SET8_DIR	DSPGP3 SET8_DIR	DSPGP2 SET8_DIF	DSPGP1 RSET8_DIR	
R315664 4D110h)	DSPGP_SET8_Level_1	0 DSPGP16	0 DSPGP15	0 DSPGP14	0 DSPGP13	0 DSPGP12	0 DSPGP11	0 DSPGP10	0 DSPGP9	0 DSPGP8	0 DSPGP7	0 DSPGP6	0 DSPGP5	0 DSPGP4	0 DSPGP3	0 DSPGP2	0 DSPGP1	00000000
,		_SET8_ _LVL	SET8_LVL	SET8_LVL	SET8_LVL	SET8_LVL	SET8_LVL	SET8_LVL	SET8_LVL	SET8_LVĪ	DSPGP1 SET8_LVL							
R524288 (80000h)	DSP1_PMEM_0	0	0	0	0	0	0	0	0 SP1 PM S	START [31:1	6]	C	SP1_PM_S	START [39:3	32]			00000000
R524290 80002h)	DSP1_PMEM_1	0	0	0	0	0	0		DSP1_PM_				DSP1 PM	1 [39:32]				00000000
,	DSP1_PMEM_2		Ū	Ū	Ū	Ū	Ū	Ū	DSP1_PN	A_1 [31:16]			0011_1	<u> </u>				00000000
R561146 88FFAh)	DSP1_PMEM_18429	0	0	0	0	0	0	0	0 DSP1_PI 0 DSP1_PM	M_1 [15:0]	61	[DSP1_PM_	12286 [39:3	2]			00000000
R561148 88FFCh)	DSP1_PMEM_18430	0	0	0	0	0	0		DSP1_PM_ 0				DSP1_PM_	END [39:32	2]			00000001
R561150 88FFEh)	DSP1_PMEM_18431		•	•	•	•	•	•	DSP1_PM_ DSP1_PM	END [31:16 _END [15:0]								00000000
R655360 A0000h)	DSP1_XMEM_0	0	0	0	0	0	0	0	0 DSP1 XM 3	START [15:	01	C	SP1_XM_S	START [23:1	16]			00000000
R655362 A0002h)	DSP1_XMEM_1	0	0	0	0	0	0	0	0	M_1 [15:0]			DSP1_XN	A_1 [23:16]				00000000
A9FFCh)	DSP1_XMEM_20478	0	0	0	0	0	0	0	0 DSP1_XM_	20478 [15:0)]		DSP1_XM_2					00000001
R696318 A9FFEh)	DSP1_XMEM_20479	0	0	0	0	0	0	0	0 DSP1_XM	_END [15:0]]		DSP1_XM_	END [23:16	6]			00000000
R786432 C0000h)	DSP1_YMEM_0	0	0	0	0	0	0	0	0 DSP1_YM_	START [15:	0]	C	SP1_YM_S	START [23:1	16]			00000001
R786434 C0002h)	DSP1_YMEM_1	0	0	0	0	0	0	0	0 DSP1 YI	M 1 [15:0]	·		DSP1_YM	A_1 [23:16]				00000000
R802812 C3FFCh)	DSP1_YMEM_8190	0	0	0	0	0	0	0	0 DSP1 YM	8190 [15:0]	1		DSP1_YM_	8190 [23:16	6]			00000000
R802814 C3FFEh)	DSP1_YMEM_8191	0	0	0	0	0	0	0	0	END [15:0]			DSP1_YM_	END [23:16	6]			00000000
R917504 (E0000h)	DSP1_ZMEM_0	0	0	0	0	0	0	0	0			C	SP1_ZM_S	Start [23:1	16]			00000000
. ,	DSP1_ZMEM_1	0	0	0	0	0	0	0	DSP1_ZM_ 0 DSP1_Z	M_1 [15:0]	UJ		DSP1_ZN	1_1 [23:16]				00000000
,	DSP1_ZMEM_4094	0	0	0	0	0	0	0	0	4094 [15:0]	1		DSP1_ZM_	4094 [23:16	6]			00000000
R925694 E1FFEh)	DSP1_ZMEM_4095	0	0	0	0	0	0	0	0	END [15:0]	-		DSP1_ZM_	END [23:16	6]			00000000
R1048064 (FFE00h)	DSP1_Config_1	0	0	0 DSP1 R	0 RATE [3:0]	0	0	0	0	0	0	0	0 DSP1	0 DSP1	0	0 DSP1	0 DSP1	00000000
0011		U		00°1_K	vni iz [0:0]		U	U	0	0	0	U	MEM_ENA		U	CORE_ ENA	START	
R1048066 FFE02h)	DSP1_Config_2	0	0	0	0	0	0	0 DS	0 P1 CLK FF	0 REQ_SEL [1	0 15:01	0	0	0	0	0	0	00000000
R1048068 FFE04h)	DSP1_Status_1	DSP1_ PING_ FULL	DSP1_ PONG_ FULL	0	0	0	0	0	0		10.0]	DSP1_W	DMA_ACTI	VE_CHANN	NELS [7:0]			00000000
4040070		0	0	0	0	0	0	0		0	0	0	0	0	0	0	0	00000000
R1048070 (FFE06h)	DSP1_Status_2	0	0	0	0	0	0	DSP1_DU	ALMEM_CC	0 0	0 0	0	0	0	0	0	DSP1_ CLK_	00000001
R1048072 (FFE08h)	DSP1_Status_3	0	0	0	0	0	0	0 DS	0 P1 CLK FF		0	0	0	0	0	0	AVAIĒ 0	00000000
. ,	DSP1 Watchdog 1	0	0	0	0	0	0	0		0	0	0	0	0	0	0	0	00000000

(FFE0Ah)

R1048074 DSP1_Watchdog_1

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

0

DSP1_WDT_MAX_COUNT [3:0]

0

0

0

DSP1_ WDT_ENA



Register	Name	31 15	30 14	29 13	28 12	27 11	26 10	25 9	24 8	23 7	22 6	21 5	20 4	19 3	18 2	17 1	16 0	Default
R1048080 (FFE10h)	DSP1_WDMA_Buffer_1	DSP1_START_ADDRESS_WDMA_BUFFER_1 [15:0] DSP1_START_ADDRESS_WDMA_BUFFER_0 [15:0]							00000000h									
R1048082 (FFE12h)	DSP1_WDMA_Buffer_2	DSP1_START_ADDRESS_WDMA_BUFFER_3[15:0] DSP1_START_ADDRESS_WDMA_BUFFER_2[15:0]							00000000h									
R1048084 (FFE14h)	DSP1_WDMA_Buffer_3	DSP1_START_ADDRESS_WDMA_BUFFER_5 [15:0] DSP1_START_ADDRESS_WDMA_BUFFER_4 [15:0]						00000000h										
R1048086 (FFE16h)	DSP1_WDMA_Buffer_4		DSP1_START_ADDRESS_WDMA_BUFFER_7 [15:0]							00000000h								
R1048096 (FFE20h)	DSP1_RDMA_Buffer_1		DSP1_START_ADDRESS_WDMA_BUFFER_6[15:0] DSP1_START_ADDRESS_RDMA_BUFFER_1[15:0] DSP1_START_ADDRESS_RDMA_RUFEEE_0[15:0]							00000000h								
、 ,	DSP1_RDMA_Buffer_2		DSP1_START_ADDRESS_RDMA_BUFFER_0[15:0] DSP1_START_ADDRESS_RDMA_BUFFER_3[15:0] DSP1_START_ADDRESS_RDMA_BUFFER_2[15:0]							00000000h								
· · ·	DSP1_RDMA_Buffer_3		DSP1_START_ADDRESS_RDMA_BUFFER_2[15:0] DSP1_START_ADDRESS_RDMA_BUFFER_5[15:0] DSP1_START_ADDRESS_RDMA_BUFFER_4[15:0]								00000000h							
· · ·	DSP1_DMA_Config_1	0	0	0	0	0	0	0	0		ER LENG	DSP1_W	/DMA_CHA	NNEL_EN/	ABLE [7:0]			00000000h
. ,	DSP1_DMA_Config_2	0	0	0	0	0	0	0	0		0	0	0	0	0	0	0	00000000h
. ,	DSP1_DMA_Config_3	0	0	0	0	0	0	0	0	0	0	DSP1_W	DSP1_F		NNEL_OFF			00000000h
()	DSP1 DMA Config 4	0	0	0	0	0	0	0	0	0	0	0	0		NNEL_ENA	BLE [5:0] 0	0	00000000h
(FFE36h)	3_	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	DSP1_ DMA_ WORD_ SEL	
R1048120 (FFE38h)	DSP1_External_Start	0	0	0	0	0	0	0	0	0	0	0	0	0 DSP1	0 START IN 3	0 SEL [4:0]	0	00000000h
R1048128 (FFE40h)	DSP1_Scratch_1								SP1_SCRA									00000000h
R1048130 (FFE42h)	DSP1_Scratch_2	DSP1_SCRATCH_3[15:0] DSP1_SCRATCH_2[15:0]						00000000h										
R1048146 (FFE52h)	DSP1_Bus_Error_Addr	0	0	0	0	0	0	0	0 D1 BUS EF			DSP1_BUS_ERR_ADDR [23:16]				00000000h		
R1048148 (FFE54h)	DSP1_Ext_window_A	DSP1 EXT_A PSIZĒ16	0	0	0	0	0	0	0 SP1 EXT /	0	0	0	0	0	0	0	0	00000000h
R1048150 (FFE56h)	DSP1_Ext_window_B	DSP1_ EXT_B	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	00000000h
(PSIZE16						D	SP1_EXT_E	B_PAGE [15	5:0]							-
R1048152 (FFE58h)	DSP1_Ext_window_C	DSP1_ EXT_C_ PSIZE16	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	00000000h
D4040454	DOD4 Fut window D	DSP1	0	0	0	0	0	D:	SP1_EXT_0	C_PAGE [1:	5:0]	0	0	0		0		00000000
(FFE5Ah)	DSP1_Ext_window_D	EXT_D PSIZE16	U	U	0	0	U					0	U	U	0	U	0	00000000h
R1048158	DSP1 Watchdog 2	0	0	0	0	0	0	D:	SP1_EXT_E	D_PAGE [1: 0	5:0] 0	0	0	0	0	0	0	00000000h
(FFE5Eh)	_ 0_		<u>^</u>		<u>^</u>				SP1_WDT_									
(FFE60h)	DSP1_Identity	0	0	0	0	0	0	0	0	0	0	0	0	0 DSP1_0	0 CORE_NUM	0 BER [4:0]	0	00000000h
R1048164 (FFE64h)	DSP1_Region_lock_sts_ 0	0	0	0	0	0	0	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	0 DSP1_	00000000h
								CTRL REGION9 LOCK_ STS	CTRL [_] REGION8_ LOCK_ STS	CTRL ⁻ REGION7 LOCK_ STS	CTRL REGION6 LOCK STS	CTRL [_] REGION5 LOCK_ STS	CTRL ⁻ REGION4 LOCK_ STS	CTRL REGION3 LOCK STS	CTRL REGION2 LOCK STS	CTRL [_] REGION1_ LOCK_ STS	CTRL [_] REGION0_ LOCK_ STS	-
R1048166 (FFE66h)	DSP1_Region_lock_1 _DSP1_Region_lock_0							00000000h										
R1048168 (FFE68h)	DSP1_Region_lock_3 _DSP1_Region_lock_2	DSP1_CTRL_REGION3_LOCK [15:0] DSP1_CTRL_REGION2_LOCK [15:0]						00000000h										
R1048170 (FFE6Ah)	DSP1_Region_lock_5_ _DSP1_Region_lock_4	DSP1_CTRL_REGION5_LOCK [15:0] DSP1_CTRL_REGION4_LOCK [15:0]						00000000h										
R1048172 (FFE6Ch)	DSP1_Region_lock_7 _DSP1_Region_lock_6							00000000h										
R1048174 (FFE6Eh)	DSP1_Region_lock_9 DSP1_Region_lock_8					00000000h												
. ,		0 DSP1_ LOCK_ ERR_STS	0 DSP1_ ADDR_ ERR_STS	0 DSP1_ WDT_ TIMEOUT_	0	0	0	0	0	0	0	0	0	0	0	0 DSP1_ ERR_ PAUSE	0 DSP1_ ERR_ CLEAR	00000000h
(FFE7Ch)	DSP1_PMEM_Err_ AddrXMEM_Err_ Addr	0		STS		<u> </u>	<u> </u>	DSP	DSP1_PM 1_XMEM_E		ADDR [14:0] [15:0]			<u> </u>	<u> </u>	<u> </u>	<u> </u>	00000000h



7 Thermal Characteristics

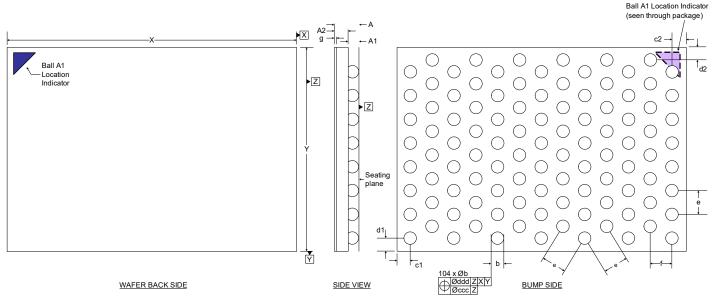
Parameter	Symbol	WLCSP	Units
Junction-to-ambient thermal resistance	θ_{JA}	38.5	°C/W
Junction-to-board thermal resistance	θ_{JB}	9.3	°C/W
Junction-to-case thermal resistance	θJC	2.11	°C/W
Junction-to-board thermal-characterization parameter	Ψ _{JB}	9.3	°C/W
Junction-to-package-top thermal-characterization parameter	Ψ _{JT}	0.10	°C/W

Notes:

- Natural convection at the maximum recommended operating temperature T_A (see Table 3-3)
- Four-layer, 2s2p PCB as specified by JESD51-9 and JESD51-11; dimensions: 101.5 x 114.5 x 1.6 mm
- Thermal parameters as defined by JESD51-12



8 Package Dimensions



Notes:

- Dimensioning and tolerances per ASME Y 14.5M–2009.
- The Ball A1 position indicator is for illustration purposes only and may not be to scale.
- Dimension "b" applies to the solder sphere diameter and is measured at the midpoint between the package body and the seating plane Datum Z.

Dimension	Millimeters							
Dimension	Minimum	Nominal	Maximum					
A	0.464	0.494	0.524					
A1	0.161	0.19	0.219					
A2	0.289	0.304	0.319					
b	0.24	0.27	0.3					
c1	0.3279	0.3579	0.3879					
d1	0.2062	0.2362	0.2662					
c2	0.2307	0.2607	0.2907					
d2	0.206	0.236	0.266					
е	BSC	0.4	BSC					
f	BSC	0.3464	BSC					
g	REF	0.3464	BSC					
Х	4.7504	4.7754	4.8004					
Y	3.4472	3.4722	3.4972					
ccc = 0.05	•		•					
ddd = 0.15								

Table 8-1. WLCSP Package Dimensions

Note: Controlling dimension is millimeters.

9 Ordering Information

Table 9-1. Ordering Information

Product	Description	Package	Halogen Free	Pb Free	Grade	Temperature Range	Container	Order #
CS42L92	32-Bit 384-kHz Hi-Fi Audio Codec	104-ball WLCSP	Yes	Yes	Commercial	–40 to +85°C	Tape and Reel ¹	CS42L92-CWZR

1.Reel quantity = 6000 units.



10 References

- MIPI Alliance, MIPI Alliance Specification for Serial Low-Power Inter-Chip Media Bus (SLIMbus). http:// www.mipi.org/
- Google Inc, Android Wired Headset Specification, Version 1.1. http://source.android.com/accessories/ headset-spec.html
- International Electrotechnical Commission, IEC60958-3 Digital Audio Interface—Consumer. http://www.ansi.org/

11 Revision History

Revision	Changes
F1 NOV '17	 AVDD maximum rating amended (Table 3-2). DSP memory sizes amended (Table 4-29). Package dimension tolerances added (Section 8).

Contacting Cirrus Logic Support

For all product questions and inquiries, contact a Cirrus Logic Sales Representative.

To find the one nearest you, go to www.cirrus.com.

IMPORTANT NOTICE

The products and services of Cirrus Logic International (UK) Limited; Cirrus Logic, Inc.; and other companies in the Cirrus Logic group (collectively either "Cirrus Logic" or "Cirrus") are sold subject to Cirrus Logic's terms and conditions of sale supplied at the time of order acknowledgment, including those pertaining to warranty, indemnification, and limitation of liability. Software is provided pursuant to applicable license terms. Cirrus Logic reserves the right to make changes to its products and specifications or to discontinue any product or service without notice. Customers should therefore obtain the latest version of relevant information from Cirrus Logic to verify that the information is current and complete. Testing and other quality control techniques are utilized to the extent Cirrus Logic deems necessary. Specific testing of all parameters of each device is not necessarily performed. In order to minimize risks associated with customer applications, assistance or customer product design. The customer is solely responsible for its selection and use of Cirrus Logic products. Use of Cirrus Logic products may entail a choice between many different modes of operation, some or all of which may require action by the user, and some or all of which may be optional. Nothing in these materials should be interpreted as instructions or suggestions to choose one mode over another. Likewise, description of a single mode should not be interpreted as a suggestion that other modes should not be used or that they would not be suitable for operation. Features and operations described herein are for illustrative purposes only.

CERTAIN APPLICATIONS USING SEMICONDUCTOR PRODUCTS MAY INVOLVE POTENTIAL RISKS OF DEATH, PERSONAL INJURY, OR SEVERE PROPERTY OR ENVIRONMENTAL DAMAGE ("CRITICAL APPLICATIONS"). CIRRUS LOGIC PRODUCTS ARE NOT DESIGNED, AUTHORIZED OR WARRANTED FOR USE IN PRODUCTS SURGICALLY IMPLANTED INTO THE BODY, AUTOMOTIVE SAFETY OR SECURITY DEVICES, NUCLEAR SYSTEMS, LIFE SUPPORT PRODUCTS OR OTHER CRITICAL APPLICATIONS. INCLUSION OF CIRRUS LOGIC PRODUCTS IN SUCH APPLICATIONS IS UNDERSTOOD TO BE FULLY AT THE CUSTOMER'S RISK AND CIRRUS LOGIC DISCLAIMS AND MAKES NO WARRANTY, EXPRESS, STATUTORY OR IMPLIED, INCLUDING THE IMPLIED WARRANTIES OF MERCHANTABILITY AND FITNESS FOR PARTICULAR PURPOSE, WITH REGARD TO ANY CIRRUS LOGIC PRODUCT THAT IS USED IN SUCH A MANNER. IF THE CUSTOMER OR CUSTOMER'S CUSTOMER USES OR PERMITS THE USE OF CIRRUS LOGIC PRODUCTS IN CRITICAL APPLICATIONS, CUSTOMER AGREES, BY SUCH USE, TO FULLY INDEMNIFY CIRRUS LOGIC, ITS OFFICERS, DIRECTORS, EMPLOYEES, DISTRIBUTORS AND OTHER AGENTS FROM ANY AND ALL LIABILITY, INCLUDING ATTORNEYS' FEES AND COSTS, THAT MAY RESULT FROM OR ARISE IN CONNECTION WITH THESE USES.

This document is the property of Cirrus Logic and by furnishing this information, Cirrus Logic grants no license, express or implied, under any patents, mask work rights, copyrights, trademarks, trade secrets or other intellectual property rights. Any provision or publication of any third party's products or services does not constitute Cirrus Logic's approval, license, warranty or endorsement thereof. Cirrus Logic gives consent for copies to be made of the information contained herein only for use within your organization with respect to Cirrus Logic integrated circuits or other products of Cirrus Logic, and only if the reproduction is without alteration and is accompanied by all associated copyright, proprietary and other notices and conditions (including this notice). This consent does not extend to other copying such as copying for general distribution, advertising or promotional purposes, or for creating any work for resale. This document and its information is provided "AS IS" without warranty of any kind (express or implied). All statutory warranties and conditions are excluded to the fullest extent possible. No responsibility is assumed by Cirrus Logic of the use of information herein, including use of this information as the basis for manufacture or sale of any items, or for infringement of patents or other rights of third parties. Cirrus Logic, Cirrus Logic. Other brand and product names may be trademarks or service marks of their respective owners.

Copyright © 2017 Cirrus Logic, Inc. All rights reserved.

MIPI and SLIMbus are trademarks or registered trademarks of MIPI Alliance, Inc.

USB-C is a trademark of USB Implementers Forum.

Android is a trademark of Google, Inc.

SPI is a trademark of Motorola.