TLV320ADC3001 Low-Power Stereo ADC With Embedded miniDSP for Wireless Handsets and Portable Audio

1 Features

- Stereo Audio ADC
  - 92-dBA Signal-to-Noise Ratio
  - Supports ADC Sample Rates From 8 kHz to 96 kHz
- Instruction-Programmable Embedded miniDSP
- Flexible Digital Filtering With RAM Programmable Coefficient, Instructions, and Built-In Standard Modes
  - Low-Latency IIR Filters for Voice
  - Linear Phase FIR Filters for Audio
  - Additional Programmable IIR Filters for EQ, Noise Cancellation, or Reduction
  - Up to 128 Programmable ADC Digital Filter Coefficients
- Three Audio Inputs With Configurable Automatic Gain Control (AGC)
  - Programmable in Single-Ended or Fully Differential Configurations
  - Can Be Driven Hi-Z for Easy Interoperability With Other Audio ICs
- Low Power Consumption and Extensive Modular Power Control:
  - 6-mW Mono Record 8-kHz
  - 11-mW Stereo Record, 8-kHz
  - 10-mW Mono Record, 48-kHz
  - 17-mW Stereo Record, 48-kHz
- Programmable Microphone Bias
- Programmable PLL for Clock Generation
- I^2C Control Bus
- Audio Serial Data Bus Supports I^2S, Left/Right-Justified, DSP, PCM, and TDM Modes
- Power Supplies:
  - Analog: 2.6 V–3.6 V.
  - Digital: Core: 1.65 V–1.95 V, I/O: 1.1 V–3.6 V
- 2.24-mm × 2.16-mm NanoFree™ 16-Ball 16-YZH Wafer Chip Scale Package (WCSP)

2 Applications

- Wireless Handsets
- Portable Low-Power Audio Systems
- Noise Cancellation Systems
- Front-End Voice or Audio Processor for Digital Audio

3 Description

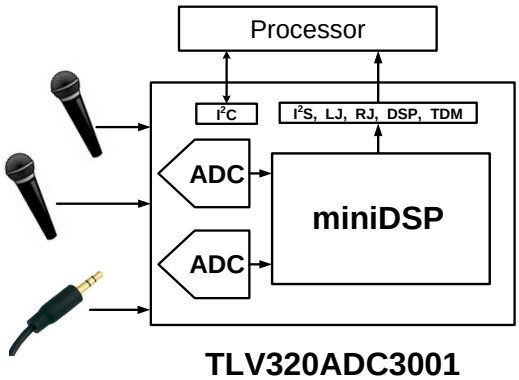
The TLV320ADC3001 device is a low-power, stereo audio analog-to-digital converter (ADC) supporting sampling rates from 8 kHz to 96 kHz with an integrated programmable-gain amplifier providing up to 40-dB analog gain or AGC. A programmable miniDSP is provided for custom audio processing. Front-end input coarse attenuation of 0 dB, –6 dB, or off, is also provided. The inputs are programmable in a combination of single-ended or fully differential configurations. Extensive register-based power control is available via I^2C, enabling mono or stereo recording. Low power consumption makes the TLV320ADC3001 ideal for battery-powered portable equipment.

Device Information(1)

PART NUMBER	PACKAGE	BODY SIZE (NOM)
TLV320ADC3001	DSBGA (16)	2.24 mm × 2.16 mm

(1) For all available packages, see the orderable addendum at the end of the data sheet.

Functional Block Diagram



TLV320ADC3001

An IMPORTANT NOTICE at the end of this data sheet addresses availability, warranty, changes, use in safety-critical applications, intellectual property matters and other important disclaimers. PRODUCTION DATA.

Table of Contents

1 Features ... 1
2 Applications ... 1
3 Description ... 1
4 Revision History ... 2
5 Description (continued) ... 4
6 Device Comparison Table ... 4
7 Pin Configuration and Functions ... 5
8 Specifications ... 6
8.1 Absolute Maximum Ratings ... 6
8.2 ESD Ratings ... 6
8.3 Recommended Operating Conditions ... 6
8.4 Thermal Information ... 6
8.5 Electrical Characteristics ... 7
8.6 Dissipation Ratings ... 8
8.7 I^2S/LJF/RJF Timing in Master Mode ... 8
8.8 DSP Timing in Master Mode ... 8
8.9 I^2S/LJF/RJF Timing in Slave Mode ... 9
8.10 DSP Timing in Slave Mode ... 9
8.11 Typical Characteristics ... 12
9 Parameter Measurement Information ... 12
10 Detailed Description ... 13
10.1 Overview ... 13
10.2 Functional Block Diagram ... 13
10.3 Feature Description ... 14
10.4 Device Functional Modes ... 40
10.5 Programming ... 40
10.6 Register Maps ... 42
11 Application and Implementation ... 74
11.1 Application Information ... 74
11.2 Typical Application ... 74
12 Power Supply Recommendations ... 78
13 Layout ... 79
13.1 Layout Guidelines ... 79
13.2 Layout Example ... 79
14 Device and Documentation Support ... 80
14.1 Community Resources ... 80
14.2 Trademarks ... 80
14.3 Electrostatic Discharge Caution ... 80
14.4 Glossary ... 80
15 Mechanical, Packaging, and Orderable Information ... 80

4 Revision History

NOTE: Page numbers for previous revisions may differ from page numbers in the current version.

Changes from Revision C (April 2011) to Revision D — **Page**

- Added *Pin Configuration and Functions* section, *ESD Ratings* table, *Feature Description* section, *Device Functional Modes*, *Application and Implementation* section, *Power Supply Recommendations* section, *Layout* section, *Device and Documentation Support* section, and *Mechanical, Packaging, and Orderable Information* section ... 1

Changes from Revision B (March 2010) to Revision C — **Page**

- Changed pinout diagram to top view ... 5
- Inserted missing table reference ... 75

Changes from Revision A (November, 2008) to Revision B — **Page**

- Added miniDSP to data-sheet title ... 1
- Added miniDSP bullet to the *Features* list ... 1
- Added a sentence about the miniDSP to the *Description* section ... 1
- Deleted YZH and WCSP options from the SIMPLIFIED BLOCK DIAGRAM. ... 4
- Alphabetized *Pin Functions* table ... 5
- Changed "complacence" to "compliance" in Note 2 of the Abs Max table. ... 6
- Changed θ_{JA} to $R_{\theta JA}$ ... 6
- Added AVDD = 3.3 V to *Electrical Characteristics* condition statement ... 7
- Added input common-mode voltage row to *Electrical Characteristics* table ... 7
- Added integrated noise row to *Electrical Characteristics* ... 8
- Added AVDD = 3.3 V to *Electrical Characteristics* condition statement ... 8
- Added metric dimensions to Note 1 of the DISSIPATIONS RATINGS table ... 8
- Added rise and fall times to the waveform, Figure 1 ... 10

Copyright © 2008–2015, Texas Instruments Incorporated

- Added rise and fall times to waveform, Figure 2 10
- Added rise and fall times to waveform, Figure 3 10
- Changed signs of nonzero errors in % ERROR column 24

Changes from Original (September 2008) to Revision A **Page**

- Changed Figure 4 - DSP Timing in Slave Mode. Added the WCLK text note. 11
- Removed note following the page 0 / register 94 description table 58
- Changed bit values from 1 and 2 to 0 and 1, respectively. 58
- Listed values 81 through 127 as reserved 58
- Replaced the listing of page-4 registers 64
- Added a listing for page-5 registers 69
- Changed Figure 44 Typical Connections 74

Copyright © 2008–2015, Texas Instruments Incorporated

5 Description (continued)

The AGC programs to a wide range of attack (7 ms–1.4 s) and decay (50 ms–22.4 s) times. A programmable noise gate function is included to avoid noise pumping. Low-latency IIR filters optimized for voice and telephony are available, as well as linear-phase FIR filters optimized for audio. Programmable IIR filters are also available and may be used for sound equalization, or to remove noise components. The audio serial bus can be programmed to support I^2S, left-justified, right-justified, DSP, PCM, and TDM modes. The audio bus may be operated in either master or slave mode.

A programmable integrated PLL is included for flexible clock generation and support for all standard audio rates from a wide range of available MCLKs, varying from 512 kHz to 50 MHz, including the most popular cases of 12-MHz, 13-MHz, 16-MHz, 19.2-MHz, and 19.68-MHz system clocks.

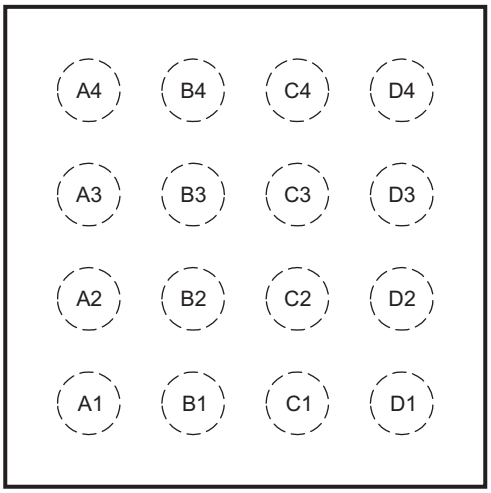
6 Device Comparison Table

FEATURES	TLV320ADC3001	TLV320ADC3101
Number of ADCs	2	2
Number of Inputs / Outputs	3 / Digital I/F	6 / Digital I/F
Resolution (Bits)	24	24
Control Interface	I^2C	I^2C
Digital Audio Interface	LJ, RJ, I2S, DSP, TDM	LJ, RJ, I2S, DSP, TDM
Digital Microphone Support	No	Yes

Copyright © 2008–2015, Texas Instruments Incorporated

7 Pin Configuration and Functions

YZH Package
16-Pin DSBGA
Top View



Pin Functions

PIN		TYPE	DESCRIPTION
NO.	NAME		
A1	MICBIAS	O	Microphone output bias voltage
A2	$\overline{\text{RESET}}$	I	Reset
A3	SCL	I/O	I^2C serial clock
A4	SDA	I/O	I^2C serial data input/output
B1	IN1R(M)	I	Analog input – first right single-ended or differential minus input
B2	AVDD	P	Analog voltage supply, 2.6 V–3.6 V
B3	DVDD	P	Digital core voltage supply, 1.65 V–1.95 V
B4	IOVDD	P	I/O voltage supply, 1.1 V–3.6 V
C1	IN1L(P)	I	Analog input – first left single-ended or differential plus input
C2	AVSS	P	Analog ground supply, 0 V
C3	DVSS	P	Digital ground supply, 0 V
C4	MCLK	I	Master clock input
D1	IN2L	I	Analog input – second left single-ended
D2	DOUT	O	Audio serial data bus data output (output)
D3	WCLK	I/O	Audio serial data bus word clock (input/output)
D4	BCLK	I/O	Audio serial data bus bit clock (input/output)

Copyright © 2008–2015, Texas Instruments Incorporated

8 Specifications

8.1 Absolute Maximum Ratings

over operating free-air temperature range (unless otherwise noted)(1)

		MIN	MAX	UNIT
	AVDD to AVSS	–0.3	3.9	V
	IOVDD to DVSS	–0.3	3.9	V
	DVDD to DVSS	–0.3	2.5	V
	Digital input voltage to DVSS	–0.3	IOVDD + 0.3	V
	Analog input voltage to AVSS	–0.3	AVDD + 0.3	V
	Operating temperature	–40	85	°C
T_J Max	Junction temperature		105	°C
	Power dissipation	$(T_J \text{ Max} – T_A) / \theta_{JA}$		W
T_{stg}	Storage temperature	–65	125	°C

(1) Stresses beyond those listed under *Absolute Maximum Ratings* may cause permanent damage to the device. These are stress ratings only, and functional operation of the device at these or any other conditions beyond those indicated under *Recommended Operating Conditions* is not implied. Exposure to absolute-maximum-rated conditions for extended periods may affect device reliability.

8.2 ESD Ratings

			VALUE	UNIT
$V_{(ESD)}$	Electrostatic discharge	Human body model (HBM), per ANSI/ESDA/JEDEC JS-001(1)	±2500	V
		Charged-device model (CDM), per JEDEC specification JESD22-C101(2)	±1000	

(1) JEDEC document JEP155 states that 500-V HBM allows safe manufacturing with a standard ESD control process.
(2) JEDEC document JEP157 states that 250-V CDM allows safe manufacturing with a standard ESD control process.

8.3 Recommended Operating Conditions

over operating free-air temperature range (unless otherwise noted)

		MIN	NOM	MAX	UNIT
AVDD(1)	Analog supply voltage	2.6	3.3	3.6	V
DVDD(1)	Digital core supply voltage	1.65	1.8	1.95	V
IOVDD(1)	Digital I/O supply voltage	1.1	1.8	3.6	V
V_I	Analog full-scale 0-dB input voltage (AVDD = 3.3 V)		0.707		Vrms
	Digital output load capacitance		10		pF
T_A	Operating free-air temperature	–40		85	°C

(1) Analog voltage values are with respect to AVSS; digital voltage values are with respect to DVSS.

8.4 Thermal Information

THERMAL METRIC(1)		TLV320ADC3001 YZH (DSBGA) 16 PINS	UNIT
$R_{\theta JA}$	Junction-to-ambient thermal resistance	70.9	°C/W
$R_{\theta JC(top)}$	Junction-to-case (top) thermal resistance	0.3	°C/W
$R_{\theta JB}$	Junction-to-board thermal resistance	13.7	°C/W
ψ_{JT}	Junction-to-top characterization parameter	1.7	°C/W
ψ_{JB}	Junction-to-board characterization parameter	13.7	°C/W
$R_{\theta JC(bot)}$	Junction-to-case (bottom) thermal resistance	—	°C/W

(1) For more information about traditional and new thermal metrics, see the *Semiconductor and IC Package Thermal Metrics* application report, SPRA953.

Copyright © 2008–2015, Texas Instruments Incorporated

8.5 Electrical Characteristics

At 25°C, AVDD = 3.3 V, IOVDD = 1.8 V, DVDD = 1.8 V, f_S = 48-kHz, 16-bit audio data (unless otherwise noted)

PARAMETER		TEST CONDITIONS	MIN	TYP	MAX	UNIT
AUDIO ADC						
	Input signal level (0-dB)	Single-ended input		0.707		Vrms
	Input common-mode voltage	Single-ended input		1.35		Vrms
	Signal-to-noise ratio, A-weighted(1) (2)	f_S = 48 kHz, 0-dB PGA gain, IN1 inputs selected and AC-shorted to ground	80	92		dB
	Dynamic range, A-weighted(1) (2)	f_S = 48 kHz, 1-kHz –60-dB full-scale input applied at IN1 inputs, 0-dB PGA gain		92		dB
THD	Total harmonic distortion	f_S = 48 kHz, 1-kHz –2-dB full-scale input applied at IN1 inputs, 0-dB PGA gain		–90	–75	dB
				0.003%	0.017%	
	Power-supply rejection ratio	234 Hz, 100 mV_{PP} on AVDD, single-ended input		46		dB
		234 Hz, 100 mV_{PP} on AVDD, differential input		68		
	ADC channel separation	1 kHz, –2 dB IN1L to IN1R		–73		dB
	ADC gain error	1-kHz input, 0-dB PGA gain		0.7		dB
	ADC programmable-gain amplifier maximum gain	1-kHz input tone, R_{SOURCE} < 50 Ω		40		dB
	ADC programmable-gain amplifier step size			0.502		dB
	Input resistance	IN1 inputs, routed to single ADC Input mix attenuation = 0 dB		35		kΩ
		IN2 inputs, input mix attenuation = 0 dB		35		
		IN1 inputs, input mix attenuation = –6 dB		62.5		
		IN2 inputs, input mix attenuation = –6 dB		62.5		
	Input capacitance			10		pF
	Input level control minimum attenuation setting			0		dB
	Input level control maximum attenuation setting			6		dB
	Input level control attenuation step size			6		dB
ADC DIGITAL DECIMATION FILTER		**f_S = 48 kHz**				
	Filter gain from 0 to 0.39 f_S	Filter A, AOSR = 128 or 64		±0.1		dB
	Filter gain from 0.55 f_S to 64 f_S	Filter A, AOSR = 128 or 64		–73		dB
	Filter group delay	Filter A, AOSR = 128 or 64		17/f_S		s
	Filter gain from 0 to 0.39 f_S	Filter B, AOSR = 64		±0.1		dB
	Filter gain from 0.60 f_S to 32 f_S	Filter B, AOSR = 64		-46		dB
	Filter group delay	Filter B, AOSR = 64		11/f_S		s
	Filter gain from 0 to 0.39 f_S	Filter C, AOSR = 32		±0.033		dB
	Filter gain from 0.28 f_S to 16 f_S	Filter C, AOSR = 32		-60		dB
	Filter group delay	Filter C, AOSR = 32		11/f_S		s
MICROPHONE BIAS						
	Bias voltage	Programmable settings, load = 750 Ω		2		V
			2.25	2.5	2.75	
				AVDD – 0.2		
	Current sourcing	2.5 V setting			4	mA

(1) Ratio of output level with 1-kHz full-scale sine-wave input, to the output level with the inputs short-circuited, measured A-weighted over a 20-Hz to 20-kHz bandwidth using an audio analyzer.
(2) All performance measurements done with 20-kHz lowpass filter and, where noted, A-weighted filter. Failure to use such a filter may result in higher THD+N and lower SNR and dynamic range readings than shown in the Electrical Characteristics. The lowpass filter removes out-of-band noise, which, although not audible, may affect dynamic specification values.

Copyright © 2008–2015, Texas Instruments Incorporated

Electrical Characteristics (continued)

At 25°C, AVDD = 3.3 V, IOVDD = 1.8 V, DVDD = 1.8 V, f_S = 48-kHz, 16-bit audio data (unless otherwise noted)

PARAMETER		TEST CONDITIONS		MIN	TYP	MAX	UNIT
	Integrated noise	BW = 20 Hz to 20 kHz, A-weighted, 1-µF capacitor between MICBIAS and AGND			3.3		µVrms
DIGITAL I/O							
V_{IL}	Input low level	I_{IL} = 5 µA		–0.3		0.3 × IOVDD	V
V_{IH}	Input high level(3)	I_{IH} = 5 µA		0.7 × IOVDD			V
V_{OL}	Output low level	I_{IH} = 2 TTL loads				0.1 × IOVDD	V
V_{OH}	Output high level	I_{OH} = 2 TTL loads		0.8 × IOVDD			V
SUPPLY CURRENT		**f_S = 48 kHz, AVDD = 3.3 V, DVDD = IOVDD = 1.8 V**					
	Mono record	AVDD	PLL and AGC off		2		mA
		DVDD			1.9		
	Stereo record	AVDD	PLL and AGC off		4		mA
		DVDD			2.1		
	PLL	AVDD	Additional power consumed when PLL is powered		1.1		mA
		DVDD			0.8		
	Power down	AVDD	All supply voltages applied, all blocks programmed in lowest power state		0.04		µA
		DVDD			0.7		

(3) When IOVDD < 1.6 V, minimum V_{IH} is 1.1 V.

8.6 Dissipation Ratings(1)

PACKAGE TYPE	T_A = 25°C POWER RATING	DERATING FACTOR	T_A = 75°C POWER RATING	T_A = 85°C POWER RATING
DSBGA	1052.6 mW	13.1 mW/°C	394.7 mW	263.2 mW

(1) This data was taken using 2 oz. (0.071-mm thick) trace and copper pad that is soldered directly to a JEDEC standard 4-layer 3-in. × 3-in. (7.62-cm × 7.62-cm) PCB.

8.7 I²S/LJF/RJF Timing in Master Mode

Specified at 25°C, DVDD = 1.8 V. All timing specifications are measured at characterization. See Figure 1 for timing diagram.

		IOVDD = 1.8 V		IOVDD = 3.3 V		UNIT
		MIN	MAX	MIN	MAX	
t_d(WS)	BCLK/WCLK delay time		20		15	ns
t_d(DO-WS)	BCLK/WCLK to DOUT delay time		25		20	ns
t_d(DO-BCLK)	BCLK to DOUT delay time		20		15	ns
t_r	Rise time		20		15	ns
t_f	Fall time		20		15	ns

8.8 DSP Timing in Master Mode

Specified at 25°C, DVDD = 1.8 V. All timing specifications are measured at characterization. See Figure 2 for timing diagram.

		IOVDD = 1.8 V		IOVDD = 3.3 V		UNIT
		MIN	MAX	MIN	MAX	
t_d(WS)	BCLK/WCLK delay time		25		15	ns
t_d(DO-BCLK)	BCLK to DOUT delay time		25		15	ns
t_r	Rise time		20		15	ns
t_f	Fall time		20		15	ns

Copyright © 2008–2015, Texas Instruments Incorporated

8.9 I²S/LJF/RJF Timing in Slave Mode

Specified at 25°C, DVDD = 1.8 V. All timing specifications are measured at characterization. See Figure 3 for timing diagram.

		IOVDD = 1.8 V		IOVDD = 3.3 V		UNIT
		MIN	MAX	MIN	MAX	
t_H(BCLK)	BCLK high period	35		35		ns
t_L(BCLK)	BCLK low period	35		35		ns
t_s(WS)	BCLK/WCLK setup time	10		6		ns
t_h(WS)	BCLK/WCLK hold time	10		6		ns
t_d(DO-WS)	BCLK/WCLK to DOUT delay time (for LJF Mode only)		30		30	ns
t_d(DO-BCLK)	BCLK to DOUT delay time		25		20	ns
t_r	Rise time		16		8	ns
t_f	Fall time		16		8	ns

8.10 DSP Timing in Slave Mode

Specified at 25°C, DVDD = 1.8 V. All timing specifications are measured at characterization. See Figure 4 for timing diagram.

		IOVDD = 1.8 V		IOVDD = 3.3 V		UNIT
		MIN	MAX	MIN	MAX	
t_H(BCLK)	BCLK high period	35		35		ns
t_L(BCLK)	BCLK low period	35		35		ns
t_s(WS)	BCLK/WCLK setup time	10		8		ns
t_h(WS)	BCLK/WCLK hold time	10		8		ns
t_d(DO-BCLK)	BCLK to DOUT delay time		25		20	ns
t_r	Rise time		15		8	ns
t_f	Fall time		15		8	ns

Copyright © 2008–2015, Texas Instruments Incorporated

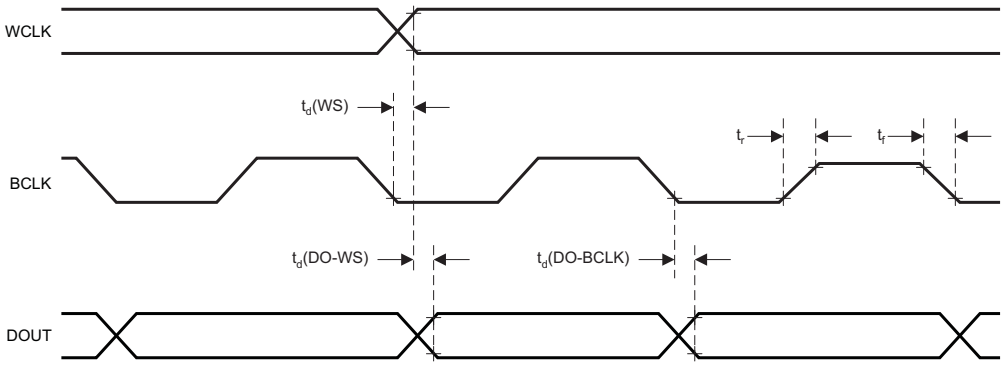


Figure 1. I²S/LJF/RJF Timing in Master Mode

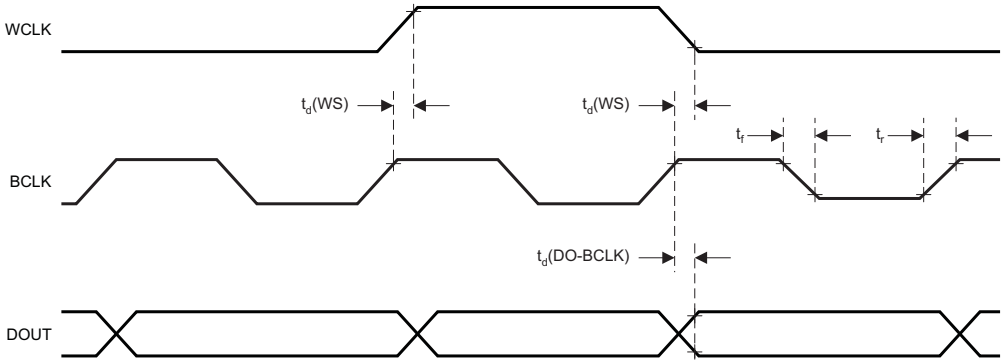


Figure 2. DSP Timing in Master Mode

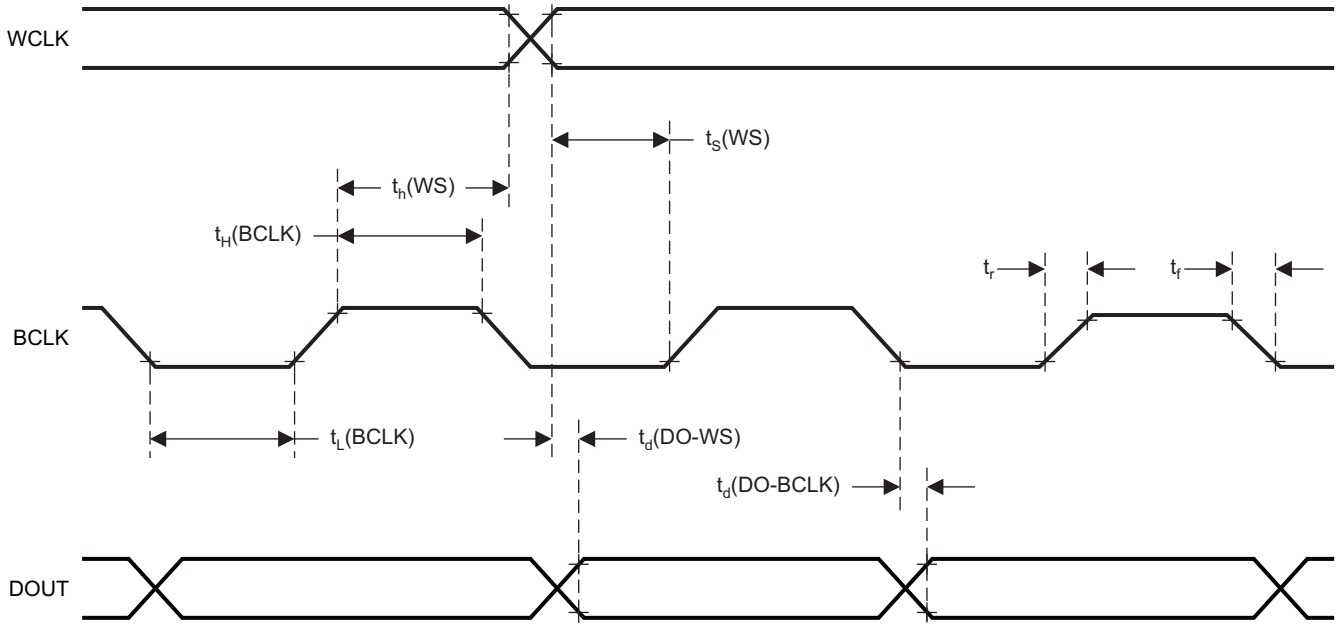
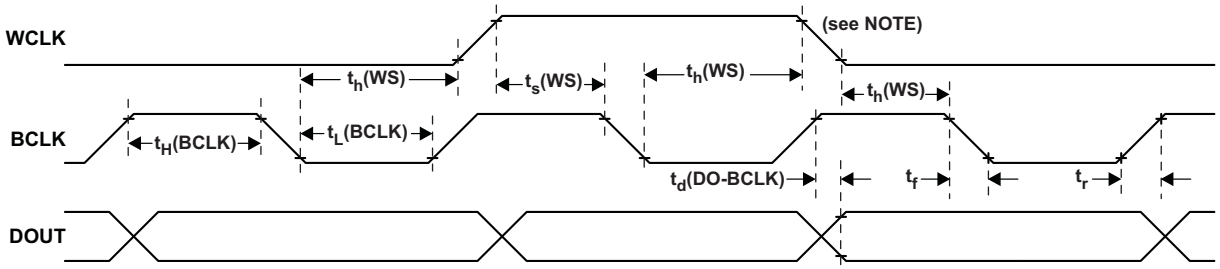


Figure 3. I²S/LJF/RJF Timing in Slave Mode

Copyright © 2008–2015, Texas Instruments Incorporated



Note A. Falling edge inside a frame for **WCLK** is arbitrary inside frame.

Figure 4. DSP Timing in Slave Mode

8.11 Typical Characteristics



Figure 5. Single-Ended Gain Error

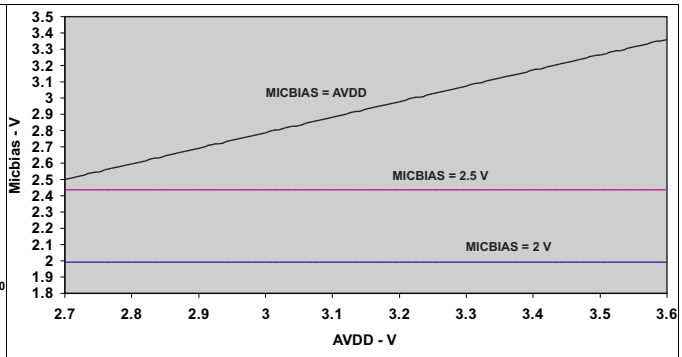


Figure 6. MICBIAS Output Voltage vs AVDD

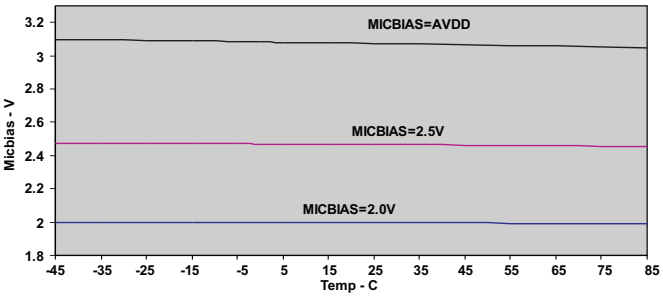


Figure 7. MICBIAS Output Voltage vs Ambient Temperature

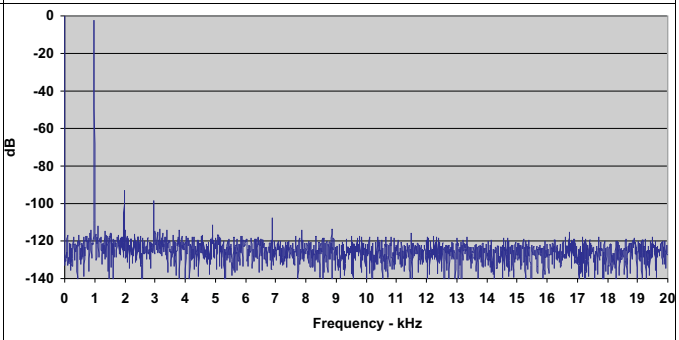


Figure 8. Line Input to ADC FFT Plot

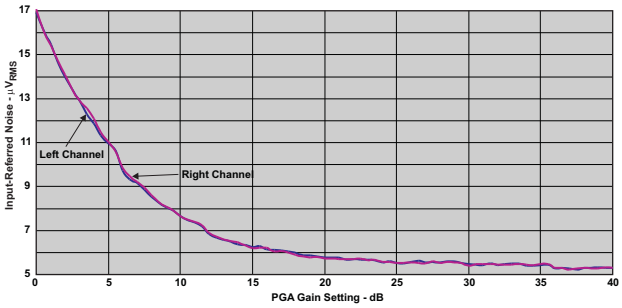


Figure 9. Input-Referred Noise vs PGA Gain

9 Parameter Measurement Information

All parameters are measured according to the conditions described in the *Specifications* section.

Copyright © 2008–2015, Texas Instruments Incorporated

10 Detailed Description

10.1 Overview

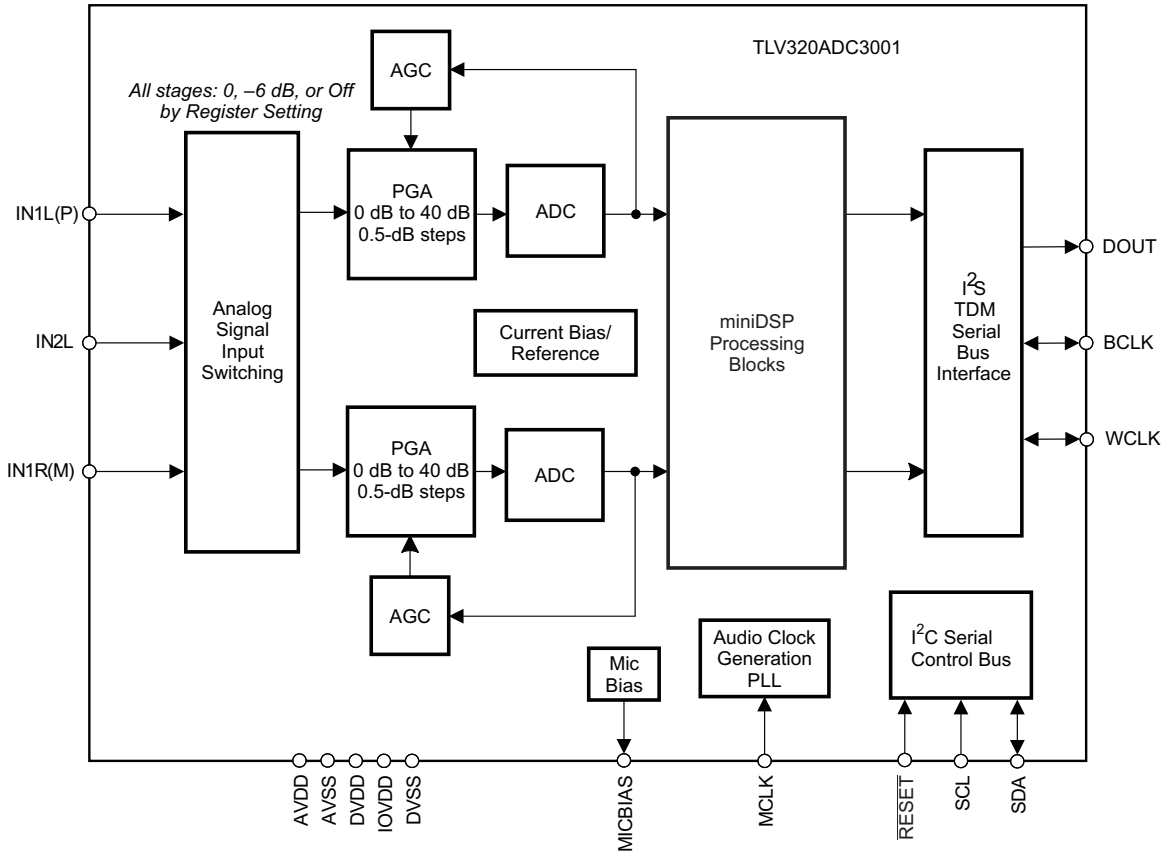
The TLV320ADC3001 is a flexible, low-power, stereo audio ADC product with extensive feature integration, intended for applications in smartphones, PDAs, and portable computing, communication, and entertainment applications. The product integrates a host of features to reduce cost, board space, and power consumption in space-constrained, battery-powered, portable applications.

The TLV320ADC3001 consists of the following blocks:

- Stereo audio multibit delta-sigma ADC (8 kHz–96 kHz)
- miniDSP for custom processing
- Built-in processing blocks for selectable digital audio effects (3-D, bass, treble, midrange, EQ, de-emphasis)
- Register configurable combinations of up to three single-ended or one differential and one single-ended audio inputs
- Fully programmable PLL with extensive ADC clock source and divider options for maximum end-system design flexibility
- 16-ball wafer chip-scale package (DSBGA YZH)

Communication to the TLV320ADC3001 for control is via a two-wire I^2C interface. The I^2C interface supports both standard and fast communication modes.

10.2 Functional Block Diagram



10.3 Feature Description

10.3.1 Hardware Reset

The TLV320ADC3001 requires a hardware reset after power up for proper operation. After all power supplies are at their specified values, the $\overline{\text{RESET}}$ pin must be driven low for at least 10 ns. If this reset sequence is not performed, the TLV320ADC3001 may not respond properly to register reads/writes.

10.3.2 PLL Start-up

When the PLL is powered on, a start-up delay of approximately 10 ms occurs after the power-up command of the PLL and before the clocks are available to the TLV320ADC3001. This delay is to ensure stable operation of the PLL and clock-divider logic.

10.3.3 Software Power Down

By default, all circuit blocks are powered down following a reset condition. Hardware power up of each circuit block can be controlled by writing to the appropriate control register. This approach allows the lowest power-supply current for the functionality required. However, when a block is powered down, all of the register settings are maintained as long as power is still being applied to the device.

10.3.4 miniDSP

The TLV320ADC3001 features a miniDSP core which is tightly coupled to the ADC. The fully programmable algorithms for the miniDSP must be loaded into the device after power up. The miniDSP has direct access to the digital stereo audio stream, offering the possibility for advanced, very low-group-delay DSP algorithms. The ADC miniDSP has 512 programmable instructions, 256 data memory locations, and 128 programmable coefficients.

Software development for the TLV320ADC3001 is supported through TI's comprehensive PurePath™ Studio software development environment, a powerful, easy-to-use tool designed specifically to simplify software development on Texas Instruments miniDSP audio platforms. The graphical development environment consists of a library of common audio functions that can be dragged and dropped into an audio signal flow and graphically connected together. The DSP code can then be assembled from the graphical signal flow with the click of a mouse. See the TLV320ADC3001 product folder on www.ti.com to learn more about PurePath Studio software and the latest status on available, ready-to-use DSP algorithms.

10.3.5 Audio Data Converters

The TLV320ADC3001 supports the following standard audio sampling rates: 8 kHz, 11.025 kHz, 12 kHz, 16 kHz, 22.05 kHz, 24 kHz, 32 kHz, 44.1 kHz, 48 kHz, 88.2 kHz, and 96 kHz. The converters can also operate at different sampling rates in various combinations, which are described further as follows.

The TLV320ADC3001 supports a wide range of options for generating clocks for the ADC section as well as the digital interface section and the other control blocks as shown in Figure 27. The clocks for the ADC require a source reference clock. The clock can be provided on device pins MCLK and BCLK. The source reference clock for the ADC section can be chosen by programming the ADC_CLKIN value on page 0 / register 4, bits D1–D0. The ADC_CLKIN can then be routed through highly flexible clock dividers shown in Figure 27 to generate various clocks required for the ADC and programmable digital filter sections. In the event that the desired audio or programmable digital filter clocks cannot be generated from the external reference clocks on MCLK and BCLK, the TLV320ADC3001 also provides the option of using an on-chip PLL, which supports a wide range of fractional multiplication values to generate the required system clocks. Starting from ADC_CLKIN, the TLV320ADC3001 provides for several programmable clock dividers to help achieve a variety of sampling rates for the ADC and the clocks for the programmable digital filter section.

10.3.6 Digital Audio Data Serial Interface

Audio data is transferred between the host processor and the TLV320ADC3001 via the digital-audio serial-data interface, or audio bus. The audio bus on this device is flexible, including left- or right-justified data options, support for I^2S or PCM protocols, programmable data-length options, a TDM mode for multichannel operation, flexible master/slave configurability for each bus clock line, and the ability to communicate with multiple devices within a system directly.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

The audio serial interface on the TLV320ADC3001 has an extensive I/O control to allow for communicating with two independent processors for audio data. The processors can communicate with the device one at a time. This feature is enabled by register programming of the various pin selections.

The audio bus of the TLV320ADC3001 can be configured for left- or right-justified, I^2S, DSP, or TDM modes of operation, where communication with standard telephony PCM interfaces is supported within the TDM mode. These modes are all MSB-first, with data width programmable as 16, 20, 24, or 32 bits by configuring page 0 / register 27, bits D5–D4. In addition, the word clock and bit clock can be independently configured in either master or slave mode for flexible connectivity to a wide variety of processors. The word clock is used to define the beginning of a frame, and may be programmed as either a pulse or a square-wave signal. The frequency of this clock corresponds to the maximum of the selected ADC sampling frequencies.

The bit clock is used to clock in and out the digital audio data across the serial bus. When in master mode, this signal can be programmed to generate variable clock pulses by controlling the bit-clock divider in page 0 / register 30 (see Figure 27). Accommodating various word lengths as well as supporting the case when multiple TLV320ADC3001s share the same audio bus may require that the number of bit-clock pulses in a frame be adjusted.

The TLV320ADC3001 also includes a feature to offset the position of the start of data a transfer with respect to the word clock. There are two configurations that allow the user to use either a single offset for both channels or to use separate offsets. Ch_Offset_1 reference represents the value in page 0 / register 28, and Ch_Offset_2 represents the value in page 0 / register 37. When page 0 / register 38, bit D0 is set to zero (time-slot-based channel assignment is disabled), the offset of both channels is controlled, in terms of number of bit clocks, by the programming in page 0 / register 28 (Ch_Offset_1). When page 0 / register 38, bit D0 = 1 (time-slot-based channel assignment enabled), the first channel is controlled, in terms of number of bit clocks, by the programming in page 0 / register 28 (Ch_Offset_1), and the second channel is controlled, in terms of number of bit clocks, by the programming in page 0 / register 37 (Ch_Offset_2), where register 37 programs the delay between the first word and the second word. Also, the relative order of the two channels can be swapped, depending on the programmable register bit (page 0 / register 38, bit D4) that enables swapping of the channels.

The TLV320ADC3001 also supports a feature of inverting the polarity of the bit clock used for transferring the audio data as compared to the default clock polarity used. This feature can be used independently of the mode of audio interface chosen. This can be configured by writing to page 0 / register 29, bit D3.

The TLV320ADC3001 further includes programmability (page 0 / register 27, bit D0) to place DOUT in the high-impedance state at the end of data transfer (that is, at the end of the bit cycle corresponding to the LSB of a channel). By combining this capability with the ability to program at what bit clock in a frame the audio data begins, time-division multiplexing (TDM) can be accomplished, resulting in multiple ADCs able to use a single audio serial data bus. To further enhance the 3-state capability, the TLV320ADC3001 can be put in a high-impedance state half of a bit cycle earlier by setting page 0 / register 38, bit D1 to 1. When the audio serial data bus is powered down while configured in master mode, the pins associated with the interface are put into a high-impedance output state.

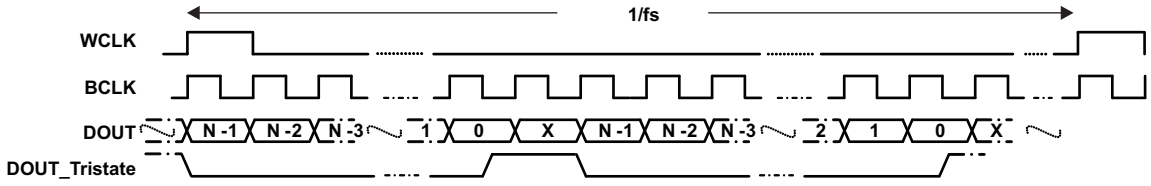


Figure 10. Both Channels Enabled, Early Hi-Z State Enabled

Either or both of the two channels can be disabled in LJF, I^2S, and DSP modes by using page 0 / register 38, bits D3–D2. Figure 10 shows the interface timing when both channels are enabled and early Hi-Z state is enabled. Figure 11 shows the effect of setting page 0 / register 38, bit D2, first channel disabled, and setting page 0 / register 27, bit D0 to 1, which enables placing DOUT in the high-impedance state. If placing DOUT in the high-impedance state is disabled, then the DOUT signal is driven to logic level 0.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

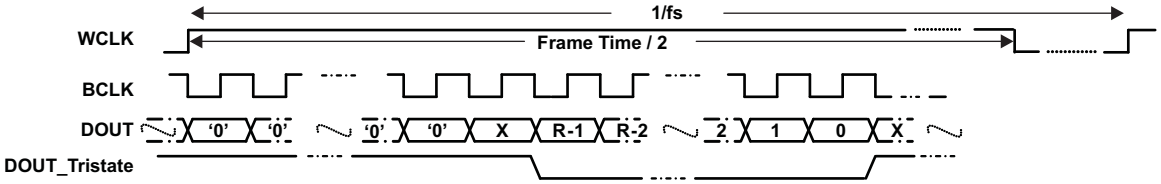


Figure 11. First Channel Disabled, Second Channel Enabled, Hi-Z State Enabled

The sync signal for the ADC filter is not generated based on the disabled channel. The sync signal for the filter corresponds to the beginning of the earlier of the two channels. If the first channel is disabled, the filter sync is generated at the beginning of the second channel, if it is enabled. If both the channels are disabled, there is no output to the serial bus, and the filter sync corresponds to the beginning of the frame.

By default, when the word clocks and bit clocks are generated by the TLV320ADC3001, these clocks are active only when the ADC is powered up within the device. This is done to save power. However, it also supports a feature wherein both the word clocks and bit clocks can be active even when the codec in the device is powered down. This is useful when using the TDM mode with multiple codecs on the same bus or when word clocks or bit clocks are used in the system as general-purpose clocks.

10.3.6.1 Right-Justified Mode

In right-justified mode, the LSB of the left channel is valid on the rising edge of the bit clock preceding the falling edge of word clock. Similarly, the LSB of the right channel is valid on the rising edge of the bit clock preceding the rising edge of the word clock. See Figure 12 for right-justifed mode timing.

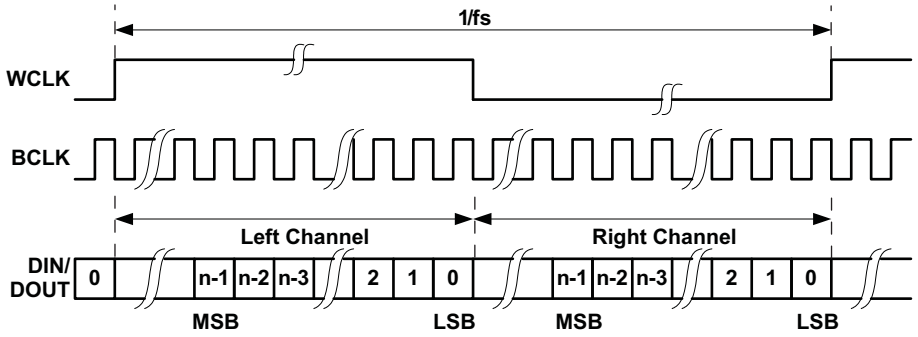


Figure 12. Timing Diagram for Right-Justified Mode

For right-justified mode, the number of bit clocks per frame must be greater than twice the programmed word length of the data.

NOTE

The time-slot-based mode is not available in the right-justified mode.

10.3.6.2 Left-Justified Mode

In left-justified mode, the MSB of the right channel is valid on the rising edge of the bit clock following the falling edge of the word clock. Similarly, the MSB of the left channel is valid on the rising edge of the bit clock following the rising edge of the word clock. Figure 13 shows the standard timing of the left-justified mode.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

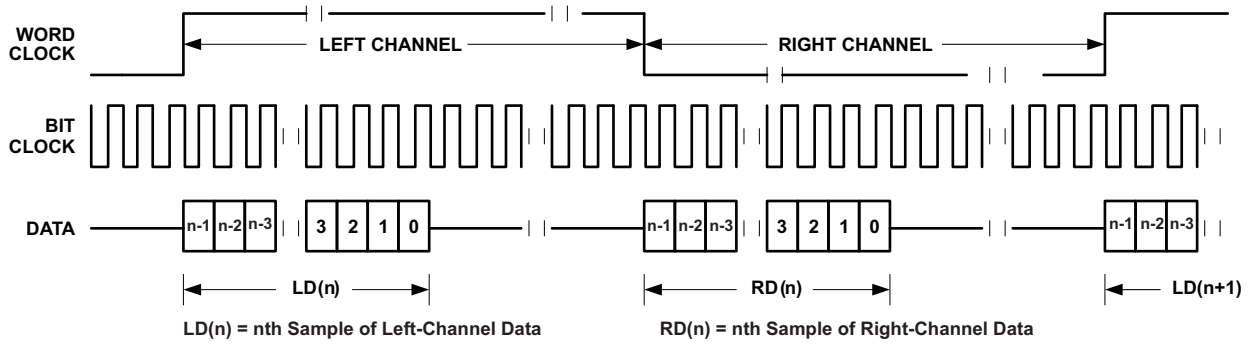


Figure 13. Left-Justified Mode (Standard Timing)

Figure 14 shows the left-justified mode with Ch_Offset_1 = 1.

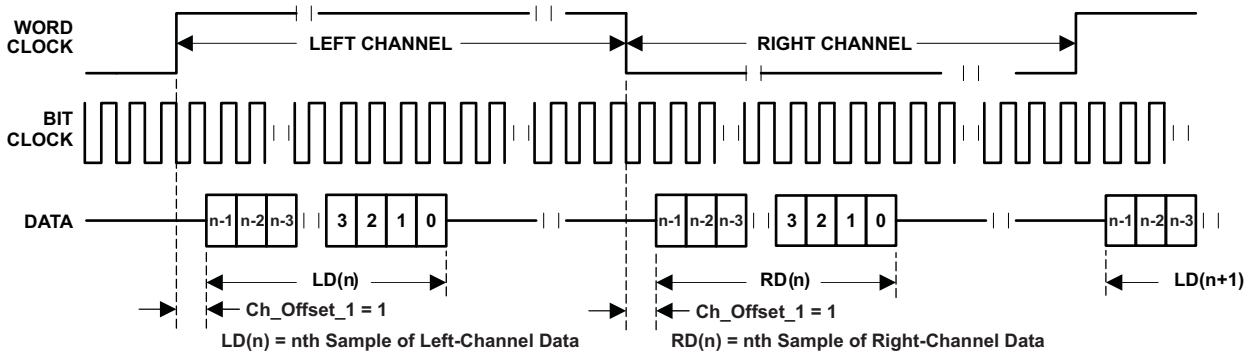


Figure 14. Left-Justified Mode With Ch_Offset_1 = 1

Figure 15 shows the left-justified mode with Ch_Offset_1 = 0 and bit clock inverted.

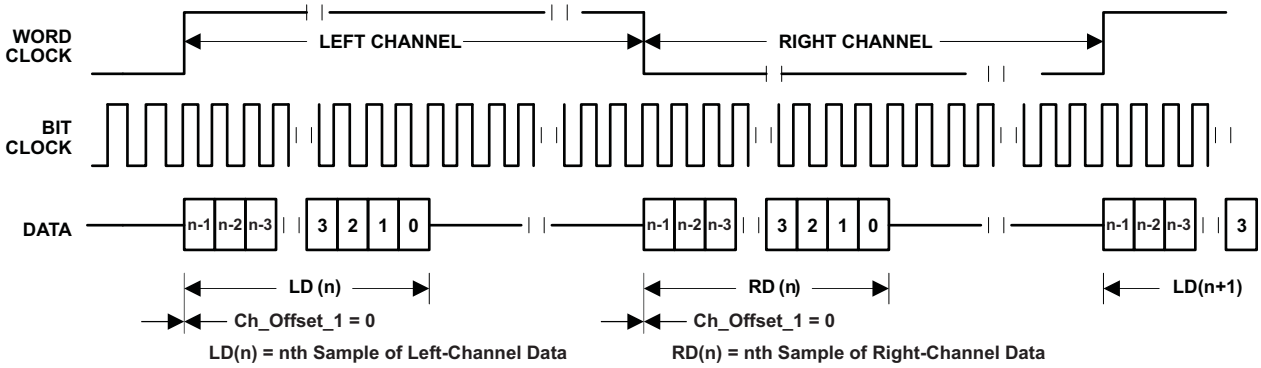


Figure 15. Left-Justified Mode With Ch_Offset_1 = 0, Bit Clock Inverted

For left-justified mode, the number of bit clocks per frame must be greater than twice the programmed word length of the data. Also, the programmed offset value must be less than the number of bit clocks per frame by at least the programmed word length of the data.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

When the time-slot-based channel assignment is disabled (page 0 / register 38, bit D0 = 0), the left and right channels have the same offset Ch_Offset_1 (page 0 / register 28), and each edge of the word clock starts data transfer for one of the two channels, depending on whether or not channel swapping is enabled. Data bits are valid on the rising edges of the bit clock. With the time-slot-based channel assignment enabled (page 0 / register 38, bit D0 = 1), the left and right channels have independent offsets (Ch_Offset_1 and Ch_Offset_2). The rising edge of the word clock starts data transfer for the first channel after a delay of its programmed offset (Ch_Offset_1) for this channel. Data transfer for the second channel starts after a delay of its programmed offset (Ch_Offset_2) from the LSB of the first-channel data. The falling edge of the word clock is not used.

With no channel swapping, the MSB of the left channel is valid on the (Ch_Offset_1 + 1)th rising edge of the bit clock following the rising edge of the word clock. And, the MSB of the right channel is valid on the (Ch_Offset_1 + 1)th rising edge of the bit clock following the falling edge of the word clock. The operation in this case, with offset of 1, is shown in the timing diagram of Figure 14. Because channel swapping is not enabled, the left-channel data is before the right-channel data. With channel swapping enabled, the MSB of the right channel is valid on the (Ch_Offset_1 + 1)th rising edge of the bit clock following the rising edge of the word clock. And, the MSB of the left channel is valid on the (Ch_Offset_1 + 1)th rising edge of the bit clock following the falling edge of the word clock. The operation in this case, with offset of 1, is shown in the timing diagram of Figure 16. As shown in the diagram, the right-channel data of a frame is before the left-channel data of that frame, due to channel swapping. Otherwise, the behavior is similar to the case where channel swapping is disabled. The MSB of the right-channel data is valid on the second rising edge of the bit clock after the rising edge of the word clock, due to an offset of 1. Similarly, the MSB of the left-channel data is valid on the second rising edge of the bit clock after the falling edge of the word clock.

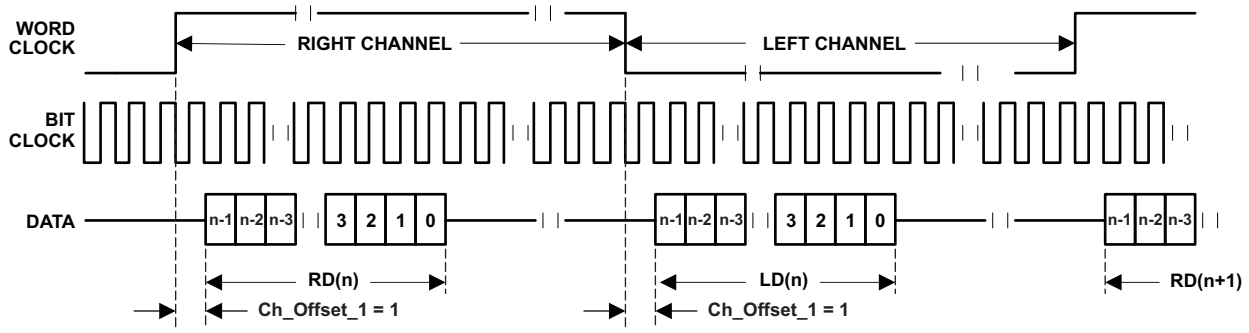


Figure 16. Left-Justified Mode With Ch_Offset_1 = 1, Channel Swapping Enabled

When time-slot-based mode is enabled with no channel swapping, the MSB of the left channel is valid on the (Offset_1 + 1)th rising edge of the bit clock following the rising edge of the word clock. And, the MSB of the right channel is valid on the (Ch_Offset_2 + 1)th rising edge of the bit clock following the LSB of the left channel.

Figure 17 shows the operation with time-slot-based mode enabled and Ch_Offset_1 = 0 and Ch_Offset_2 = 1. The MSB of the left channel is valid on the first rising edge of the bit clock after the rising edge of the word clock. Data transfer for the right channel does not wait for the falling edge of the word clock, and the MSB of the right channel is valid on the second rising edge of the bit clock after the LSB of the left channel.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

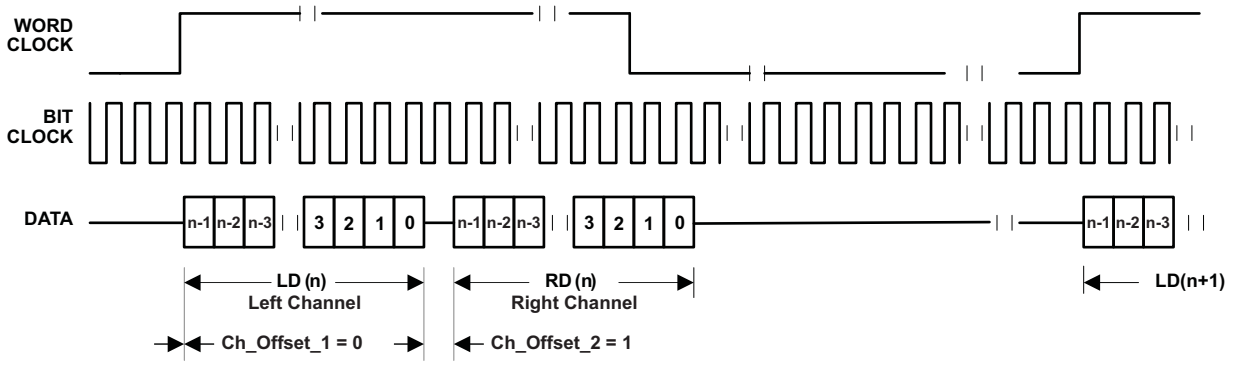


Figure 17. Left-Justified Mode, Time-Slot-Based Mode Enabled, Ch_Offset_1 = 0, Ch_Offset_2 = 1

For the case with time-slot-based mode enabled and channel swapping enabled, the MSB of the right channel is valid on the (Ch_Offset_1 + 1)th rising edge of the bit clock following the rising edge of the word clock. And, the MSB of the left channel is valid on the (Ch_Offset_2 + 1)th rising edge of the bit clock following the LSB of the right channel. Figure 18 shows the operation in this mode with Ch_Offset_1 = 0 and Ch_Offset_2 = 1. The MSB of the right channel is valid on the first rising edge of the bit clock after the rising edge of the word clock. Data transfer for the left channel starts following the completion of data transfer for the right channel without waiting for the falling edge of the word clock. The MSB of the left channel is valid on the second rising edge of the bit clock after the LSB of the right channel.

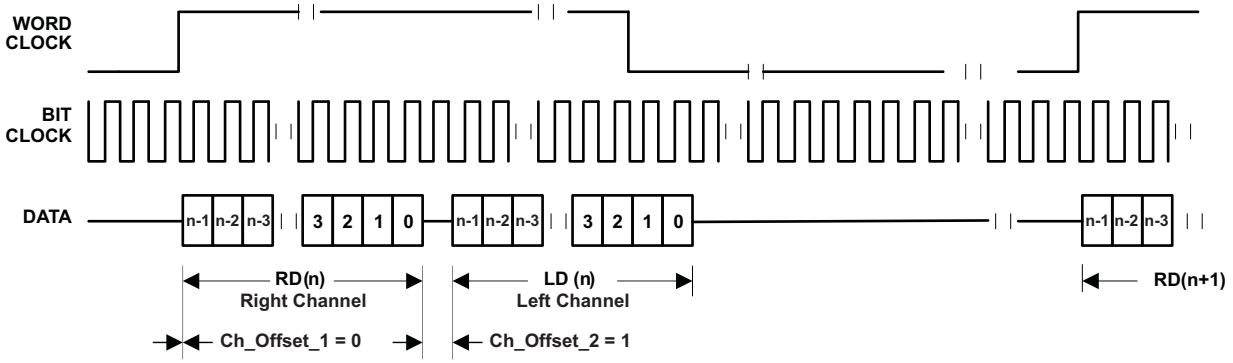


Figure 18. Left-Justified Mode, Time-Slot-Based Mode Enabled, Ch_Offset_1 = 0, Ch_Offset_2 = 1, Channel Swapping Enabled

10.3.6.3 I^2S Mode

In I^2S mode, the MSB of the left channel is valid on the second rising edge of the bit clock after the falling edge of the word clock. Similarly, the MSB of the right channel is valid on the second rising edge of the bit clock after the rising edge of the word clock. Figure 19 shows the standard I^2S timing.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

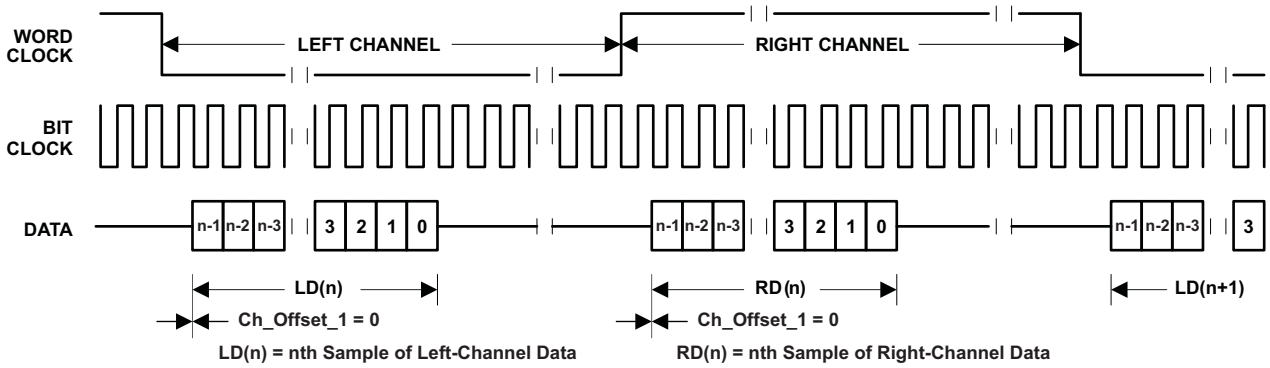


Figure 19. I²S Mode (Standard Timing)

Figure 20 shows the I²S mode timing with Ch_Offset_1 = 2.

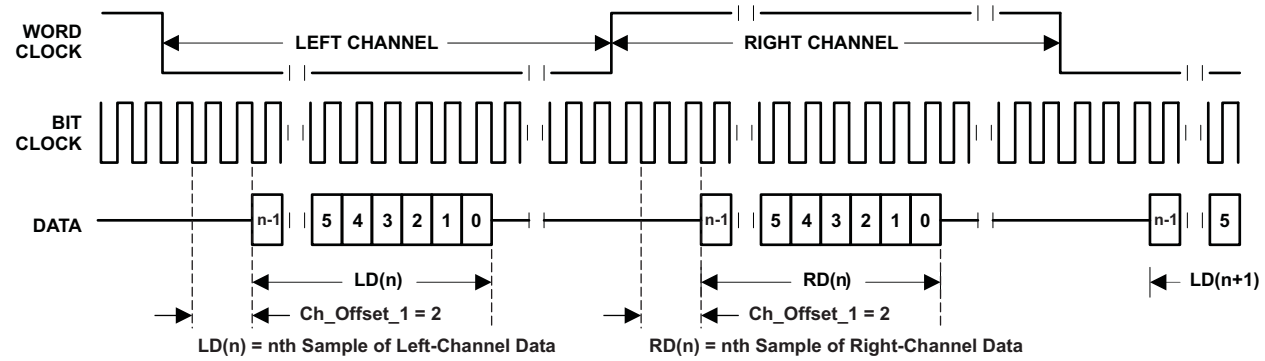


Figure 20. I²S Mode With Ch_Offset_1 = 2

Figure 21 shows the I²S mode timing with Ch_Offset_1 = 0 and bit clock inverted.

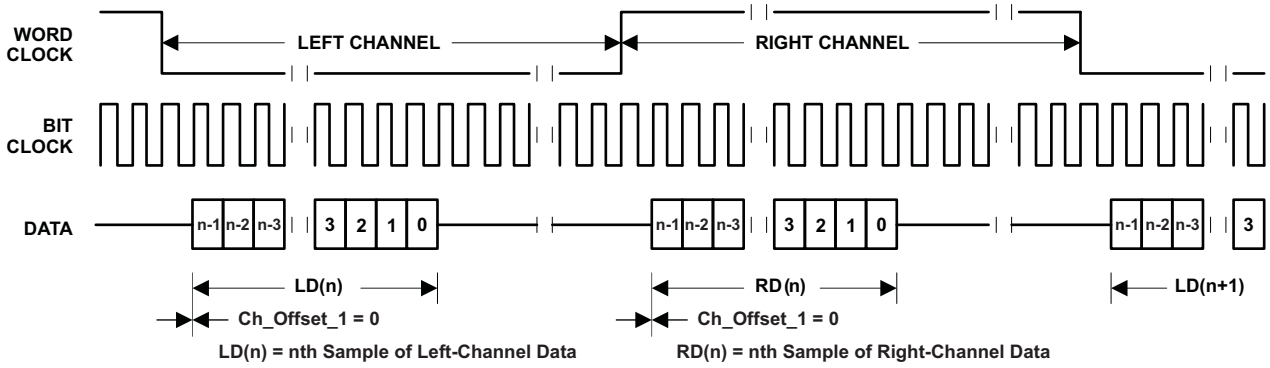


Figure 21. I²S Mode With Ch_Offset_1 = 0, Bit Clock Inverted

For I²S mode, the number of bit clocks per channel must be greater than or equal to the programmed word length of the data. Also, the programmed offset value must be less than the number of bit clocks per frame by at least the programmed word length of the data.

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

10.3.6.4 DSP Mode

In DSP mode, the rising edge of the word clock starts the data transfer with the left-channel data first and is immediately followed by the right-channel data. Each data bit is valid on the falling edge of the bit clock. Figure 22 shows the standard timing for the DSP mode.

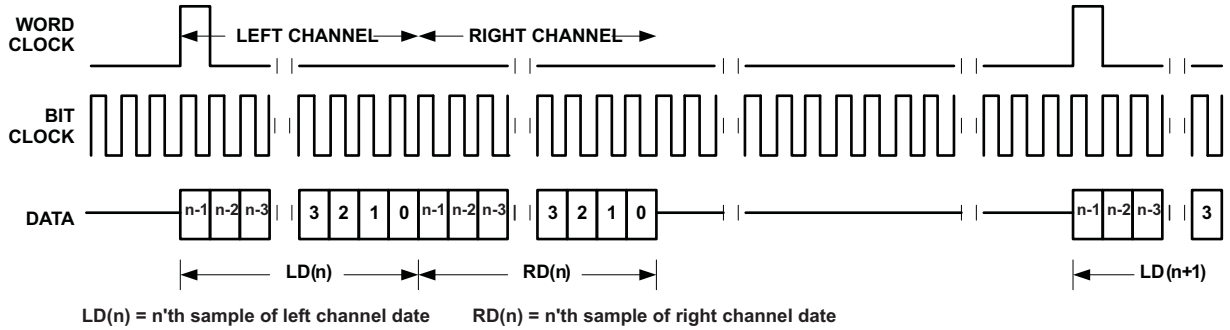


Figure 22. DSP Mode (Standard Timing)

Figure 23 shows the DSP mode timing with Ch_Offset_1 = 1.

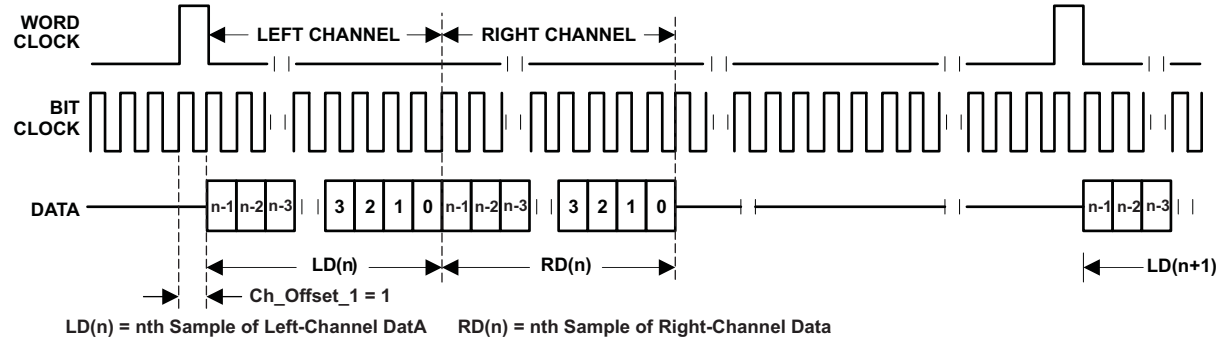


Figure 23. DSP Mode With Ch_Offset_1 = 1

Figure 24 shows the DSP mode timing with Ch_Offset_1 = 0 and bit clock inverted.

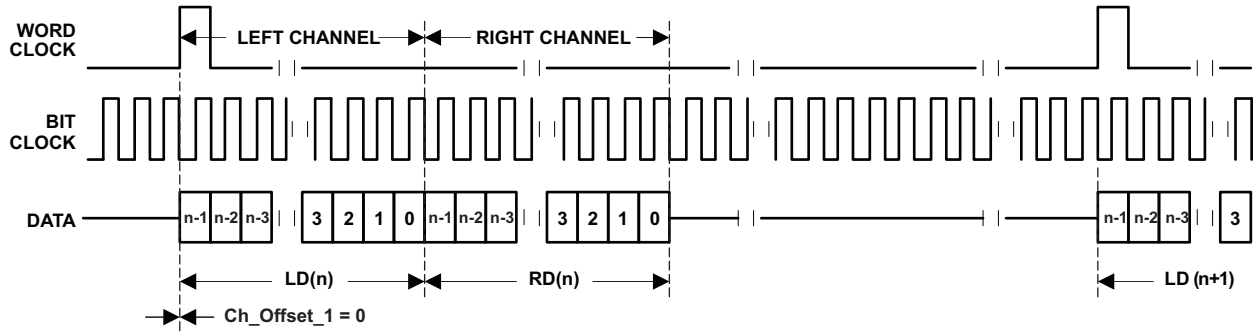


Figure 24. DSP Mode With Ch_Offset_1 = 0, Bit Clock Inverted

For DSP mode, the number of bit clocks per frame must be greater than twice the programmed word length of the data. Also, the programmed offset value must be less than the number of bit clocks per frame by at least the programmed word length of the data.

Feature Description (continued)

Figure 25 shows the DSP time-slot-based mode without channel swapping, and with Ch_Offset_1 = 0 and Ch_Offset_2 = 3. The MSB of left channel data is valid on the first falling edge of the bit clock after the rising edge of the word clock. Because the right channel has an offset of 3, the MSB of its data is valid on the third falling edge of the bit clock after the LSB of the left-channel data. As in the case of other modes, the serial output bus is put in the high-impedance state, if Hi-Z state operation of the output is enabled, during all the extra bit-clock cycles in the frame.

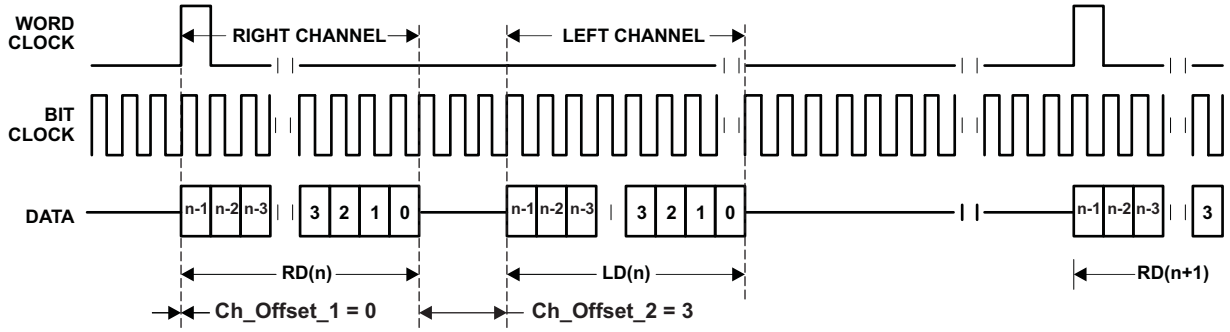


Figure 25. DSP Mode, Time-Slot-Based Mode Enabled, Ch_Offset_1 = 0, Ch_Offset_2 = 3

Figure 26 shows the timing diagram for the DSP mode with left and right channels swapped, Ch_Offset_1 = 0, and Ch_Offset_2 = 3. The MSB of the right channel is valid on the first falling edge of the bit clock after the rising edge of the word clock. And, the MSB of the left channel is valid three bit-clock cycles after the LSB of right channel, because the offset for the left channel is 3.

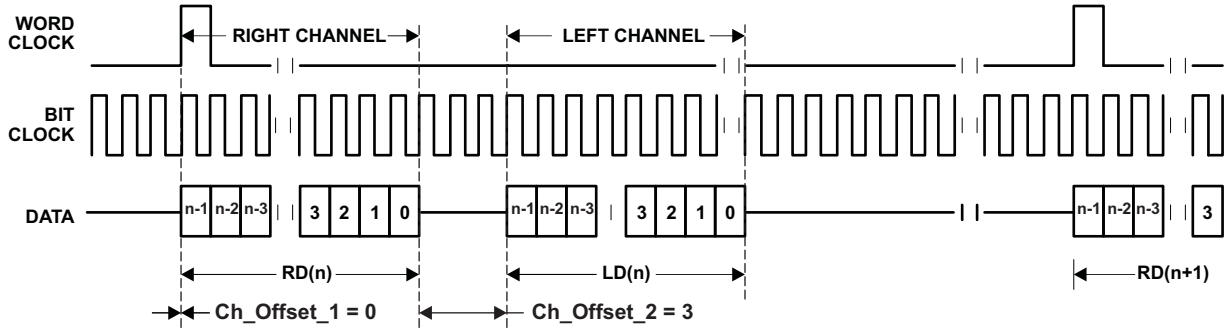


Figure 26. DSP Mode, Time-Slot-Based Mode Enabled, Ch_Offset_1 = 0, Ch_Offset_2 = 3, Channel Swap Enabled

10.3.7 Audio Clock Generation

The audio converters in fully programmable filter mode in the TLV320ADC3001 need an internal audio master clock at a frequency of ≥ N × f_S, where N = IADC (page 0, register 21) when filter mode (page 0, register 61) equals zero, otherwise N equals the instruction count from Table 6, ADC Processing Blocks. The master clock is obtained from an external clock signal applied to the device.

The device can accept an MCLK input from 512 kHz to 50 MHz, which can then be passed through either a programmable divider or a PLL, to get the proper internal audio master clock needed by the device. The BCLK input can also be used to generate the internal audio master clock.

A primary concern is proper operation of the codec at various sample rates with the limited MCLK frequencies available in the system. This device includes a highly programmable PLL to accommodate such situations easily. The integrated PLL can generate audio clocks from a wide variety of possible MCLK inputs, with particular focus paid to the standard MCLK rates already widely used.

When the PLL is enabled,

$$f_S = (PLLCLK_IN \times K \times R) / (NADC \times MADC \times AOSR \times P)$$

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

where

- P = 1, 2, 3,…, 8
- R = 1, 2, …, 16
- K = J.D
- J = 1, 2, 3, …, 63
- D = 0000, 0001, 0002, 0003, …, 9998, 9999
- PLLCLK_IN can be MCLK or BCLK, selected by page 0, register 4, bits D3–D2. (1)

P, R, J, and D are register programmable. J is the integer portion of K (the numbers to the left of the decimal point), while D is the fractional portion of K (the numbers to the right of the decimal point, assuming four digits of precision).

Examples:

If K = 8.5, then J = 8, D = 5000
If K = 7.12, then J = 7, D = 1200
If K = 14.03, then J = 14, D = 0300
If K = 6.0004, then J = 6, D = 0004

When the PLL is enabled and D = 0000, the following conditions must be satisfied to meet specified performance:

512 kHz ≤ (PLLCLK_IN / P) ≤ 20 MHz
80 MHz ≤ (PLLCLK _IN × K × R / P) ≤ 110 MHz
4 ≤ J ≤ 55

When the PLL is enabled and D ≠ 0000, the following conditions must be satisfied to meet specified performance:

10 MHz ≤ PLLCLK _IN / P ≤ 20 MHz
80 MHz ≤ PLLCLK _IN × K × R / P ≤ 110 MHz
4 ≤ J ≤ 11
R = 1

Example:

For MCLK = 12 MHz, f_S = 44.1 kHz, NADC = 8, MADC = 2, and AOSR = 128:
Select P = 1, R = 1, K = 7.5264, which results in J = 7, D = 5264

Example:

For MCLK = 12 MHz, f_S = 48 kHz , NADC = 8, MADC = 2, and AOSR = 128:
Select P = 1, R = 1, K = 8.192, which results in J = 8, D = 1920

Copyright © 2008–2015, Texas Instruments Incorporated

Feature Description (continued)

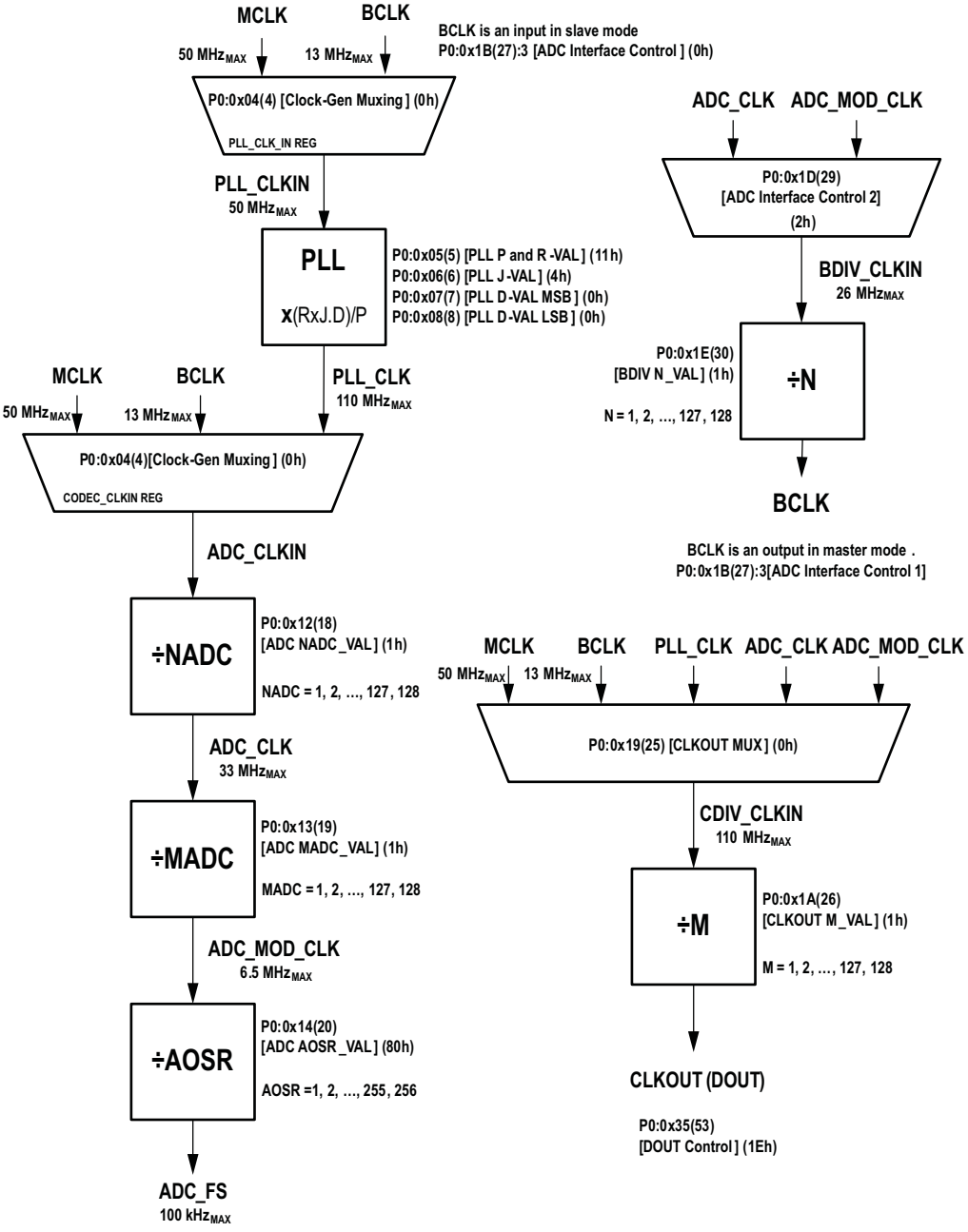
Table 1 lists several example cases of typical MCLK rates and how to program the PLL to achieve an f_S of 44.1 kHz or 48 kHz with NADC = 8, MADC = 2, and AOSR = 128.

Table 1. Typical MCLK Rates

MCLK (MHz)	P	R	J	D	ACHIEVED f_S	% ERROR
f_S = 44.1 kHz						
2.8224	1	1	32	0	44100.00	0.0000
5.6448	1	1	16	0	44100.00	0.0000
12.0	1	1	7	5264	44100.00	0.0000
13.0	1	1	6	9474	44099.71	–0.0007
16.0	1	1	5	6448	44100.00	0.0000
19.2	1	1	4	7040	44100.00	0.0000
19.68	1	1	4	5893	44100.30	0.0007
48.0	4	1	7	5264	44100.00	0.0000
f_S = 48 kHz						
2.048	1	1	48	0	48000.00	0.0000
3.072	1	1	32	0	48000.00	0.0000
4.096	1	1	24	0	48000.00	0.0000
6.144	1	1	16	0	48000.00	0.0000
8.192	1	1	12	0	48000.00	0.0000
12.0	1	1	8	1920	48000.00	0.0000
13.0	1	1	7	5618	47999.71	–0.0006
16.0	1	1	6	1440	48000.00	0.0000
19.2	1	1	5	1200	48000.00	0.0000
19.68	1	1	4	9951	47999.79	–0.0004
48.0	4	1	8	1920	48000.00	0.0000

Copyright © 2008–2015, Texas Instruments Incorporated

A detailed diagram of the audio clock section of the TLV320ADC3001 is shown in Figure 27.



Note:
MADC x AOSR ≥ IADC
Where IADC number of instructions (Instruction Count) for the ADC MAC engine, it is programmable from 2, 4, ..., 510.
Convention:
Page Number: Register Number:{Register Bit}[Register Name](Reset Value)

Figure 27. Audio Clock-Generation Processing

10.3.8 Stereo Audio ADC

The TLV320ADC3001 includes a stereo audio ADC, which uses a delta-sigma modulator with 128-times oversampling in single-rate mode, followed by a digital decimation filter. The ADC supports sampling rates from 8 kHz to 48 kHz in single-rate mode, and up to 96 kHz in dual-rate mode. Whenever the ADC is in operation, the device requires that an audio master clock be provided and appropriate audio clock generation be set up within the device.

In order to provide optimal system power dissipation, the stereo ADC can be powered one channel at a time, to support the case where only mono record capability is required. In addition, both channels can be fully or partially powered down.

The integrated digital decimation filter removes high-frequency content and downsamples the audio data from an initial sampling rate of 128 f_S to the final output sampling rate of f_S. The decimation filter provides a linear phase output response with a group delay of $17/f_S$. The –3-dB bandwidth of the decimation filter extends to 0.45 f_S and scales with the sample rate (f_S). The filter has minimum 73-dB attenuation over the stop band from 0.55 f_S to 64 f_S. Independent digital highpass filters are also included with each ADC channel, with a corner frequency that can be set independently by programmable coefficients or can be disabled entirely.

Because of the oversampling nature of the audio ADC and the integrated digital decimation filtering, requirements for analog anti-aliasing filtering are relaxed. The TLV320ADC3001 integrates a second-order analog anti-aliasing filter with 20-dB attenuation at 1 MHz. This filter, combined with the digital decimation filter, provides sufficient anti-aliasing filtering without requiring additional external components.

The ADC is preceded by a programmable-gain amplifier (PGA), which allows analog gain control from 0 dB to 40 dB in steps of 0.5 dB. The PGA gain changes are implemented with an internal soft-stepping algorithm that only changes the actual volume level by one 0.5-dB step every one or two ADC output samples, depending on the register programming (see register page 0 / register 81). This soft-stepping makes sure that volume control changes occur smoothly with no audible artifacts. On reset, the PGA gain defaults to a mute condition, and on power down, the PGA soft-steps the volume to mute before shutting down. A read-only flag is set whenever the gain applied by PGA equals the desired value set by the register. The soft-stepping control can also be disabled by programming a register bit. When soft stepping is enabled, the audio master clock must be applied to the device after the ADC power-down register is written to ensure the soft-stepping to mute has completed. When the ADC power-down flag is no longer set, the audio master clock can be shut down.

10.3.9 Audio Analog Inputs

10.3.9.1 Digital Volume Control

The TLV320ADC3001 also has a digital volume-control block with a range from –12 dB to 20 dB in steps of 0.5 dB. It is set by programming page 0 / register 83 and page 0 / register 84 for left and right channels, respectively.

Table 2. Digital Volume Control for ADC

DESIRED GAIN dB	LEFT / RIGHT CHANNEL PAGE 0 / REGISTER 83, PAGE 0 / REGISTER 84, BITS D6–D0
–12	110 1000
–11.5	110 1001
–11	110 1010
...	...
–0.5	111 1111
0	000 0000 (default)
0.5	000 0001
...	...
19.5	010 0111
20	010 1000

Copyright © 2008–2015, Texas Instruments Incorporated

During volume control changes, the soft-stepping feature is used to avoid audible artifacts. The soft-stepping rate can be set to either 1 or 2 gain steps per sample. Soft-stepping can also be entirely disabled. This soft-stepping is configured via page 0 / register 81, bits D1–D0, and is common to soft-stepping control for the analog PGA. During power-down of an ADC channel, this volume control soft-steps down to –12 dB before powering down. Due to the soft-stepping control, soon after changing the volume control setting or powering down the ADC channel, the actual applied gain may be different from the one programmed through the control register. The TLV320ADC3001 gives feedback to the user through read-only flags page 0 / register 36, bit D7 for the left channel and page 0 / register 36, bit D3 for the right channel.

10.3.9.2 Fine Digital Gain Adjustment

Additionally, the gain in each of the channels is finely adjustable in steps of 0.1 dB. This is useful when trying to match the gain between channels. By programming page 0 / register 82, the gain can be adjusted from 0 dB to –0.4 dB in steps of 0.1 dB. This feature, in combination with the regular digital volume control, allows the gains through the left and right channels be matched in the range of –0.5 dB to 0.5 dB with a resolution of 0.1 dB.

10.3.9.3 AGC

The TLV320ADC3001 includes automatic gain control (AGC) for ADC recording. AGC can be used to maintain a nominally-constant output level when recording speech. As opposed to manually setting the PGA gain, in the AGC mode, the circuitry automatically adjusts the PGA gain as the input signal becomes overly loud or weak, such as when a person speaking into a microphone moves closer to or farther from the microphone. The AGC algorithm has several programmable parameters, including target gain, attack and decay time constants, noise threshold, and max PGA applicable, that allow the algorithm to be fine-tuned for any particular application. The algorithm uses the absolute average of the signal (which is the average of the absolute value of the signal) as a measure of the nominal amplitude of the output signal. Because the gain can be changed at the sample interval time, the AGC algorithm operates at the ADC sample rate.

- **Target level** represents the nominal output level at which the AGC attempts to hold the ADC output signal level. The TLV320ADC3001 allows programming of eight different target levels, which can be programmed from –5.5 dB to –24 dB relative to a full-scale signal. Because the TLV320ADC3001 reacts to the signal absolute average and not to peak levels, it is recommended that the target level be set with enough margin to avoid clipping at the occurrence of loud sounds.
- **Attack time** determines how quickly the AGC circuitry reduces the PGA gain when the output signal level exceeds the target level due to increase in input signal level. A wide range of attack-time programmability is supported in terms of number of samples (that is, number of ADC sample-frequency clock cycles).
- **Decay time** determines how quickly the PGA gain is increased when the output signal level falls below the target level due to reduction in input signal level. A wide range of decay time programmability is supported in terms of number of samples (that is, number of ADC sample-frequency clock cycles).
- **Noise threshold** is a reference level. If the input speech average value falls below the noise threshold, the AGC considers it as a silence and hence brings down the gain to 0 dB in steps of 0.5 dB every sample period and sets the noise-threshold flag. The gain stays at 0 dB unless the input speech signal average rises above the noise threshold setting. This ensures that noise is not amplified in the absence of speech. Noise threshold level in the AGC algorithm is programmable from –30 dB to –90 dB of full-scale. When the AGC noise threshold is set to –70 dB, –80 db, or –90 dB, the microphone input *max PGA applicable* setting must be greater than or equal to 11.5 dB, 21.5 dB, or 31.5 dB, respectively. This operation includes hysteresis and debounce to avoid the AGC gain from cycling between high gain and 0 dB when signals are near the noise threshold level. The noise (or silence) detection feature can be entirely disabled by the user.
- **Max PGA applicable** allows the designer to restrict the maximum gain applied by the AGC. This can be used for limiting PGA gain in situations where environmental noise is greater than the programmed noise threshold. Microphone input Max PGA applicable can be programmed from 0 dB to 40 dB in steps of 0.5 dB.
- **Hysteresis**, as the name suggests, determines a window around the noise threshold which must be exceeded to detect that the recorded signal is indeed either noise or signal. If initially the energy of the recorded signal is greater than the noise threshold, then the AGC recognizes it as noise only when the energy of the recorded signal falls below the noise threshold by a value given by hysteresis. Similarly, after the recorded signal is recognized as noise, for the AGC to recognize it as a signal, its energy must exceed the noise threshold by a value given by the hysteresis setting. In order to prevent the AGC from jumping between noise and signal states, (which can happen when the energy of recorded signal is close to the noise threshold) a non-zero hysteresis value must be chosen. The hysteresis feature can also be disabled.
- **Debounce time (noise and signal)** determines the hysteresis in time domain for noise detection. The AGC

Copyright © 2008–2015, Texas Instruments Incorporated

continuously calculates the energy of the recorded signal. If the calculated energy is less than the set noise threshold, then the AGC does not increase the input gain to achieve the target level. However, to handle audible artifacts which can occur when the energy of the input signal is close to the noise threshold, the AGC checks if the energy of the recorded signal is less than the noise threshold for a time greater than the noise debounce time. Similarly, the AGC starts increasing the input-signal gain to reach the target level when the calculated energy of the input signal is greater than the noise threshold. Again, to avoid audible artifacts when the input-signal energy is close to noise threshold, the energy of the input signal must continuously exceed the noise threshold value for the signal-debounce time. If the debounce times are kept small, then audible artifacts can result by rapid enabling and disabling the AGC function. At the same time, if the debounce time is kept too large, then the AGC may take time to respond to changes in levels of input signals with respect to the noise threshold. Both noise and signal-debounce time can be disabled.

- The **AGC noise-threshold flag** is a read-only flag indicating that the input signal has levels lower than the noise threshold, and thus is detected as noise (or silence). In such a condition, the AGC applies a gain of 0 dB.
- **Gain applied by AGC** is a ready-only register setting which gives a real-time feedback to the system on the gain applied by the AGC to the recorded signal. This, along with the target setting, can be used to determine the input signal level. In a steady-state situation

  Target Level (dB) = Gain Applied by AGC (dB) + Input Signal Level (dB)

  When the AGC noise threshold flag is set, then the status of gain applied by AGC is not valid.
- The **AGC saturation flag** is a read-only flag indicating that the ADC output signal has not reached its target level. However, the AGC is unable to increase the gain further because the required gain is higher than the maximum allowed PGA gain. Such a situation can happen when the input signal has low energy and the noise threshold is also set low. When the AGC noise threshold flag is set, the status of AGC saturation flag must be ignored.
- The **ADC saturation flag** is a read-only flag indicating an overflow condition in the ADC channel. On overflow, the signal is clipped and distortion results. This typically happens when the AGC target level is kept high and the energy in the input signal increases faster than the attack time.
- An **AGC lowpass filter** is used to help determine the average level of the input signal. This average level is compared to the programmed detection levels in the AGC to provide the correct functionality. This lowpass filter is in the form of a first-order IIR filter. Two 8-bit registers are used to form the 16-bit digital coefficient as shown on the register map. In this way, a total of 6 registers are programmed to form the three IIR coefficients. The transfer function of the filter implemented for signal level detection is given by

$$H(z) = \frac{N_0 + N_1 z^{-1}}{2^{15} - D_1 z^{-1}} \quad (2)$$

where

- Coefficient N0 can be programmed by writing into page 4 / register 2 and page 4 / register 3.
- Coefficient N1 can be programmed by writing into page 4 / register 4 and page 4 / register 5.
- Coefficient D1 can be programmed by writing into page 4 / register 6 and page 4 / register 7.
- N0, N1, and D1 are 16-bit 2s-complement numbers and their default values implement a lowpass filter with cutoff at $0.002735 \times ADC_f_S$.

See Table 3 for various AGC programming options. AGC can be used only if the analog microphone input is routed to the ADC channel.

Table 3. AGC Parameter Settings

FUNCTION	CONTROL REGISTER LEFT ADC	CONTROL REGISTER RIGHT ADC	BIT
AGC enable	Page 0 / register 86	Page 0 / register 94	D(7)
Target Level	Page 0 / register 86	Page 0 / register 94	D(6:4)
Hysteresis	Page 0 / register 87	Page 0 / register 95	D(7:6)
Noise threshold	Page 0 / register 87	Page 0 / register 95	D(5:1)
Max PGA applicable	Page 0 / register 88	Page 0 / register 96	D(6:0)
Time constants (attack time)	Page 0 / register 89	Page 0 / register 97	D(7:0)
Time constants (decay time)	Page 0 / register 90	Page 0 / register 98	D(7:0)

Copyright © 2008–2015, Texas Instruments Incorporated

Table 3. AGC Parameter Settings (continued)

FUNCTION	CONTROL REGISTER LEFT ADC	CONTROL REGISTER RIGHT ADC	BIT
Debounce time (noise)	Page 0 / register 91	Page 0 / register 99	D(4:0)
Debounce time (signal)	Page 0 / register 92	Page 0 / register 100	D(3:0)
Gain applied by AGC	Page 0 / register 93	Page 0 / register 101	D(7:0) (read-only)
AGC noise threshold flag	Page 0 / register 45 (sticky flag), Page 0 / register 47 (non-sticky flag)	Page 0 / register 45 (sticky flag), Page 0 / register 47 (non-sticky flag)	D(6:5) (read-only)
AGC saturation flag	Page 0 / register 36 (sticky flag)	Page 0 / register 36 (sticky flag)	D(5), D(1) (read-only)
ADC saturation flag	Page 0 / register 42 (sticky flag), Page 0 / register 43 (non-sticky flag)	Page 0 / register 42 (sticky flag), Page 0 / register 43 (non-sticky flag)	D(3:2) (read-only)

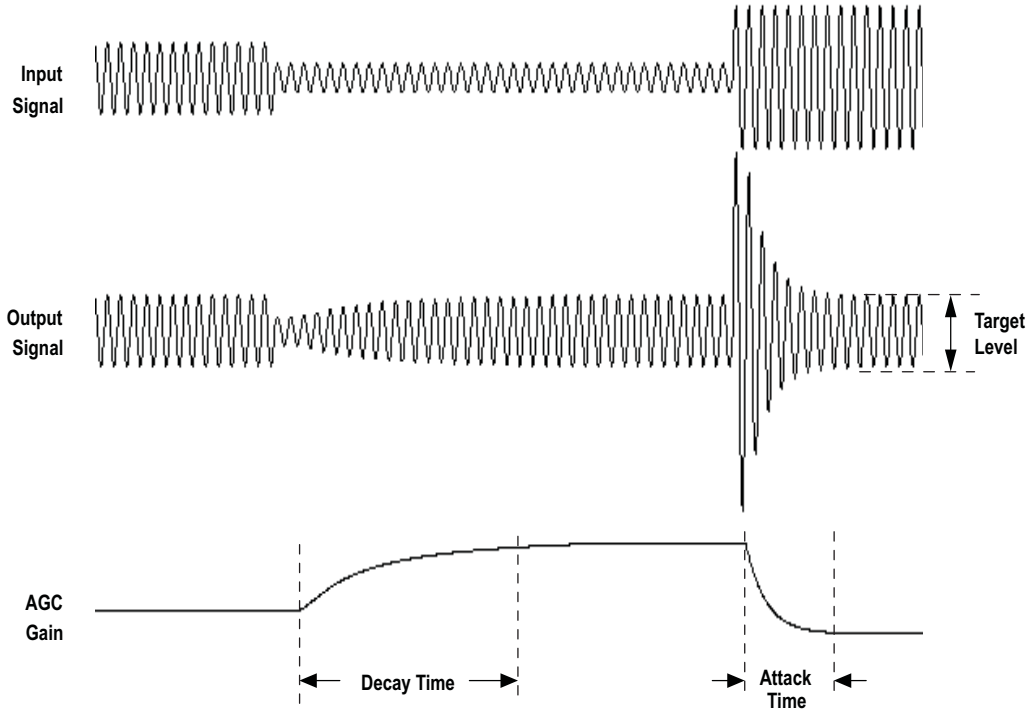


Figure 28. AGC Characteristics

The TLV320ADC3001 includes three analog audio input pins, which can be configured as one fully-differential pair and one single-ended input, or as three single-ended audio inputs. These pins connect through series resistors and switches to the virtual ground terminals of two fully differential operational amplifiers (one per ADC/PGA channel). By selecting to turn on only one set of switches per operational amplifier at a time, the inputs can be effectively multiplexed to each ADC PGA channel.

By selecting to turn on multiple sets of switches per operational amplifier at a time, mixing can also be achieved. Mixing of multiple inputs can easily lead to PGA outputs that exceed the range of the internal operational amplifiers, resulting in saturation and clipping of the mixed output signal. Whenever mixing is being implemented, the user must take adequate precautions to avoid such a saturation case from occurring. In general, the mixed signal must not exceed 2 V_{pp} (single-ended) or 4 V_{pp} (differential).

In most mixing applications, there is also a general need to adjust the levels of the individual signals being mixed. For example, if a soft signal and a large signal are to be mixed and played together, the soft signal generally must be amplified to a level comparable to the large signal before mixing. In order to accommodate this need, the TLV320ADC3001 includes input level control on each of the individual inputs before they are mixed or multiplexed into the ADC PGAs, with programmable attenuation at 0 dB, –6 dB, or off.

Copyright © 2008–2015, Texas Instruments Incorporated

NOTE

This input level control is not intended to be a volume control, but instead used for coarse level setting. Finer soft-stepping of the input level is implemented in this device by the ADC PGA.

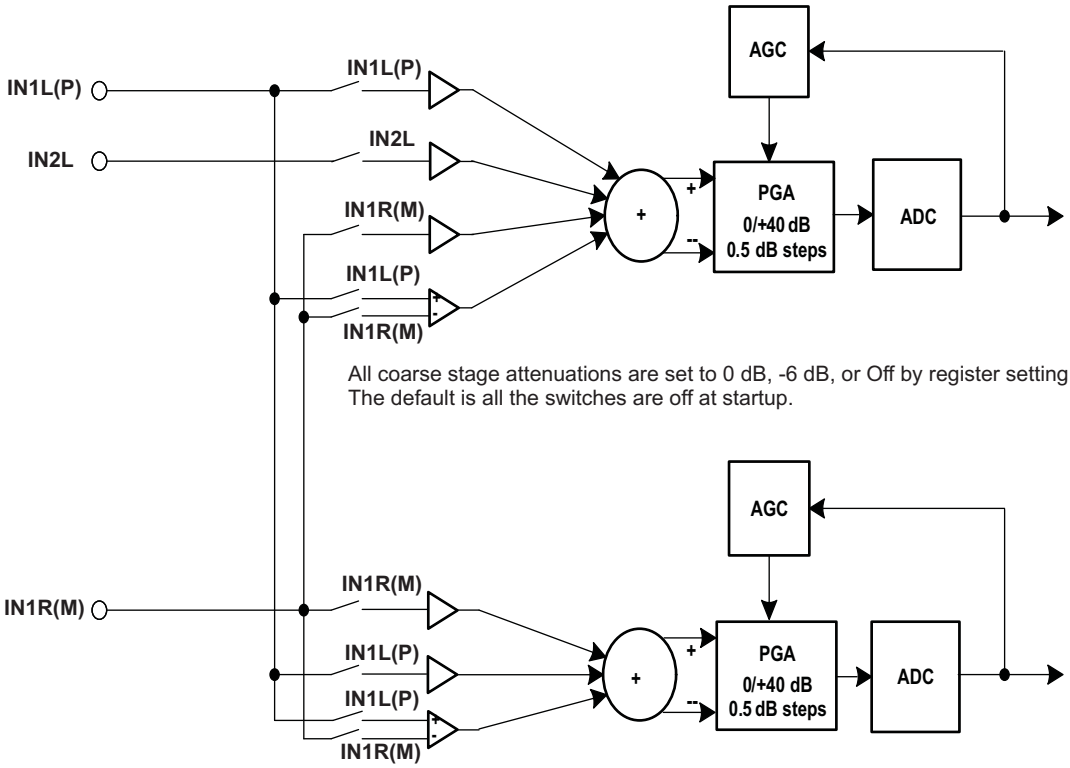


Figure 29. TLV320ADC3001 Available Audio Input Path Configurations

Table 4. TLV320ADC3001 Audio Signals

AUDIO SIGNALS AVAILABLE TO LEFT ADC		AUDIO SIGNALS AVAILABLE TO RIGHT ADC	
SINGLE-ENDED INPUTS	DIFFERENTIAL INPUTS	SINGLE-ENDED INPUTS	DIFFERENTIAL INPUTS
IN1L(P)	IN1L(P), IN1R(M)	IN1R(M)	IN1L(P), IN1R(M)
IN2L		IN1L(P)	
IN1R(M)			

Inputs can be selected as single-ended instead of fully-differential, and mixing or multiplexing into the ADC PGAs is also possible in this mode. It is not possible, however, for an input pair to be selected as fully-differential for connection to one ADC PGA and simultaneously selected as single-ended for connection to the other ADC PGA channel. However, it is possible for an input to be selected or mixed into both left- and right-channel PGAs, as long as it has the same configuration for both channels (either both single-ended or both fully differential).

10.3.10 Input Impedance and VCM Control

The TLV320ADC3001 includes several programmable settings to control analog input pins, particularly when they are not selected for connection to an ADC PGA. The default option allows unselected inputs to be put into a high-impedance state, such that the input impedance seen looking into the device is extremely high. However, the pins on the device do include protection diode circuits connected to AVDD and AVSS. Thus, if any voltage is driven onto a pin approximately one diode drop (~0.6 V) above AVDD or one diode drop below AVSS, these protection diodes will begin conducting current, resulting in an effective impedance that no longer appears as a high-impedance state.

Copyright © 2008–2015, Texas Instruments Incorporated

Another programmable option for unselected analog inputs is to weakly hold them at the common-mode input voltage of the ADC PGA (which is determined by an internal band-gap voltage reference). This is useful to keep the ac-coupling capacitors connected to analog inputs biased up at a normal dc level, thus avoiding the need for them to charge up suddenly when the input is changed from being unselected to selected for connection to an ADC PGA. This option is controlled in page 1 / register 52 through page 1 / register 57. The user must specify this option is disabled when an input is selected for connection to an ADC PGA or selected for the analog input bypass path, because it can corrupt the recorded input signal if left operational when an input is selected.

In most cases, the analog input pins on the TLV320ADC3001 must be ac-coupled to analog input sources, the only exception to this generally being if an ADC is being used for dc voltage measurement. The ac-coupling capacitor causes a highpass filter pole to be inserted into the analog signal path, so the size of the capacitor must be chosen to move that filter pole sufficiently low in frequency to cause minimal effect on the processed analog signal. The input impedance of the analog inputs when selected for connection to an ADC PGA varies with the setting of the input level control, starting at approximately 35 kΩ with an input level control setting of 0 dB, and 62.5 kΩ when the input level control is set at –6 dB. For example, using a 0.1-µF ac-coupling capacitor at an analog input will result in a highpass filter pole of 45.5 Hz when the 0-dB input level-control setting is selected. To set a highpass corner for the application, the following input impedance table (Table 5) has been provided with various mixer gains and microphone PGA ranges.

Table 5. Single-Ended Input Impedance vs PGA Ranges (1)

MIXER GAIN (dB)	MICROPHONE PGA RANGE (dB)	INPUT IMPEDANCE (Ω)
0	0–5.5	35,000
0	6–11.5	38,889
0	12–17.5	42,000
0	18–23.5	44,074
0	24–29.5	45,294
0	30–35.5	45,960
0	36–40	46,308
–6	0–5.5	62,222
–6	6–11.5	70,000
–6	12–17.5	77,778
–6	18–23.5	84,000
–6	24–29.5	88,148
–6	30–35.5	90,588
–6	36–40	91,919

(1) Valid when only one input is enabled

10.3.11 MICBIAS Generation

The TLV320ADC3001 includes a programmable microphone bias output voltage (MICBIAS), capable of providing output voltages of 2 V or 2.5 V (both derived from the on-chip band-gap voltage) with 4-mA output-current drive capability. In addition, the MICBIAS may be programmed to be switched to AVDD directly through an on-chip switch, or it can be powered down completely when not needed, for power savings. This function is controlled by register programming in page 1 / register 51.

10.3.12 ADC Decimation Filtering and Signal Processing

The TLV320ADC3001 ADC channel includes a built-in digital decimation filter to process the oversampled data from the delta-sigma modulator to generate digital data at the Nyquist sampling rate with high dynamic range. The decimation filter can be chosen from three different types, depending on the required frequency response, group delay, and sampling rate.

10.3.12.1 Processing Blocks

The TLV320ADC3001 offers a range of processing blocks which implement various signal processing capabilities along with decimation filtering. These processing blocks give users the choice of how much and what type of signal processing they may use and which decimation filter is applied.

Copyright © 2008–2015, Texas Instruments Incorporated

The signal processing blocks available are:

- First-order IIR
- Scalable number of biquad filters
- Variable-tap FIR filter
- AGC

The processing blocks are tuned for common cases and can achieve high anti-alias filtering or low group delay in combination with various signal processing effects such as audio effects and frequency shaping. The available first order IIR, biquad, and FIR filters have fully user-programmable coefficients. ADC processing blocks can be selected by writing to page 0 / register 61. The default (reset) processing block is PRB_R1.

Table 6. ADC Processing Blocks

PROCESSING BLOCKS	CHANNEL	DECIMATION FILTER	1ST ORDER IIR AVAILABLE	NUMBER BIQUADS	FIR	REQUIRED AOSR VALUE	RESOURCE CLASS
PRB_R1	Stereo	A	Yes	0	No	128, 64	6
PRB_R2	Stereo	A	Yes	5	No	128, 64	8
PRB_R3	Stereo	A	Yes	0	25-tap	128, 64	8
PRB_R4	Right	A	Yes	0	No	128, 64	3
PRB_R5	Right	A	Yes	5	No	128, 64	4
PRB_R6	Right	A	Yes	0	25-tap	128, 64	4
PRB_R7	Stereo	B	Yes	0	No	64	3
PRB_R8	Stereo	B	Yes	3	No	64	4
PRB_R9	Stereo	B	Yes	0	20-tap	64	4
PRB_R10	Right	B	Yes	0	No	64	2
PRB_R11	Right	B	Yes	3	No	64	2
PRB_R12	Right	B	Yes	0	20-tap	64	2
PRB_R13	Right	C	Yes	0	No	32	3
PRB_R14	Stereo	C	Yes	5	No	32	4
PRB_R15	Stereo	C	Yes	0	25-tap	32	4
PRB_R16	Right	C	Yes	0	No	32	2
PRB_R17	Right	C	Yes	5	No	32	2
PRB_R18	Right	C	Yes	0	25-tap	32	2

10.3.12.2 Processing Blocks – Details

10.3.12.2.1 First-Order IIR, AGC, Filter A

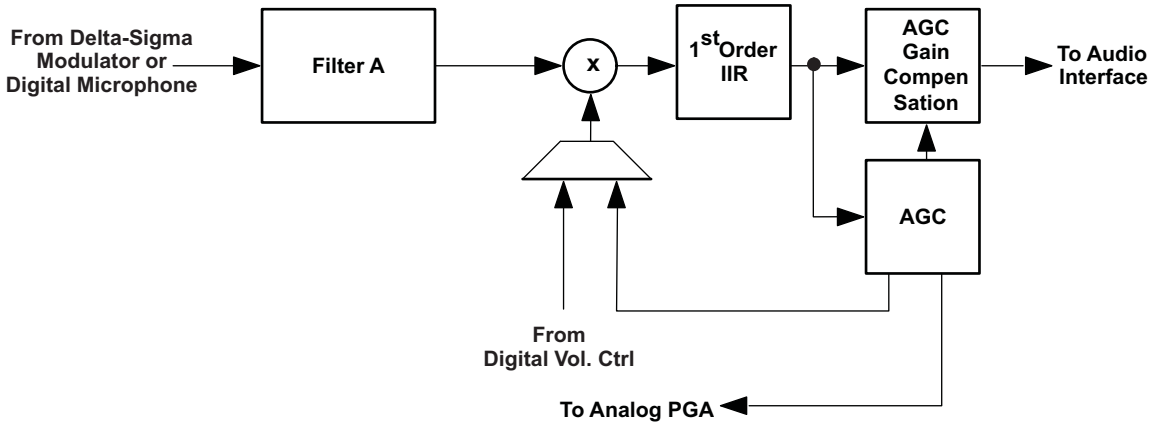


Figure 30. Signal Chain for PRB_R1 and PRB_R4

Copyright © 2008–2015, Texas Instruments Incorporated

10.3.12.2.2 Five Biquads, First-Order IIR, AGC, Filter A

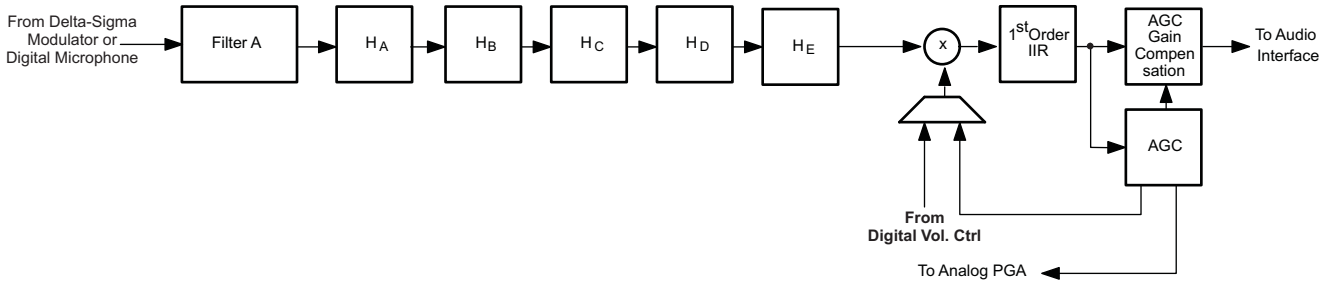


Figure 31. Signal Chain for PRB_R2 and PRB_R5

10.3.12.2.3 25-Tap FIR, First-Order IIR, AGC, Filter A

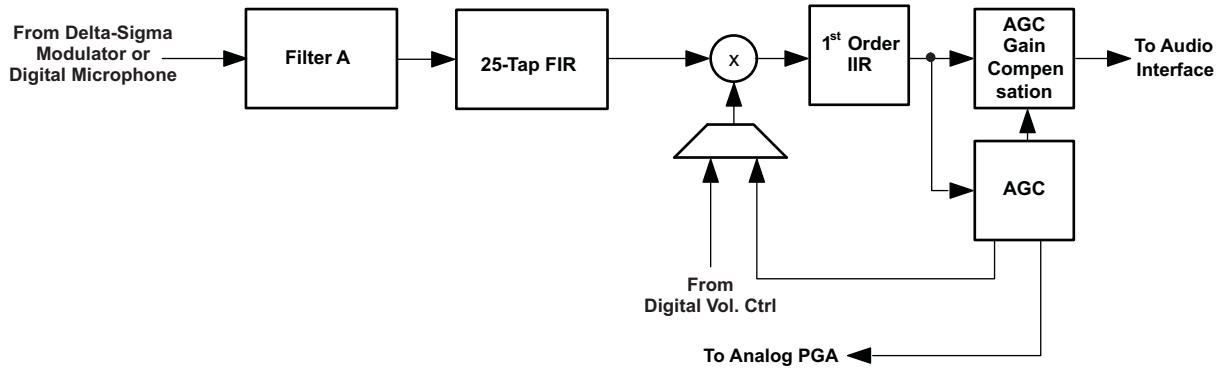


Figure 32. Signal Chain for PRB_R3 and PRB_R6

10.3.12.2.4 First-Order IIR, AGC, Filter B

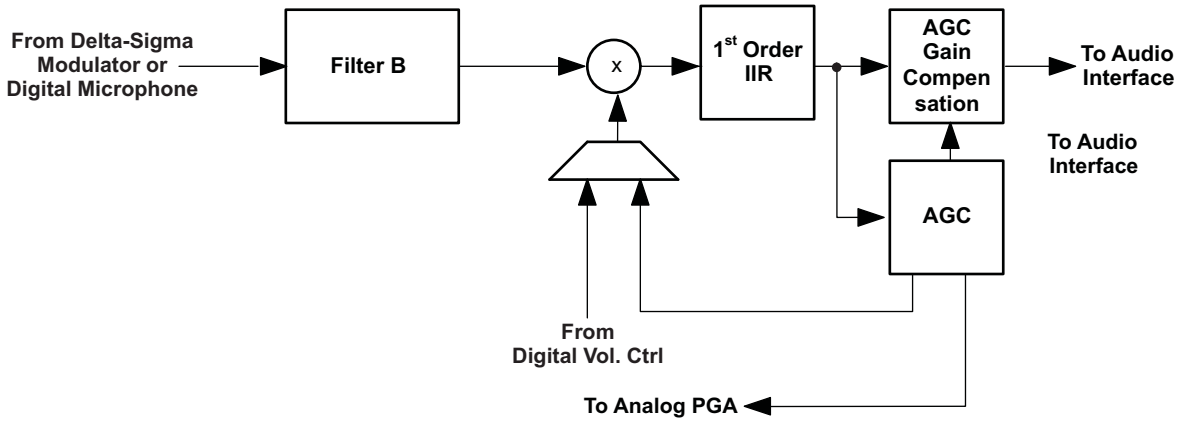


Figure 33. Signal Chain for PRB_R7 and PRB_R10

10.3.12.2.5 Three Biquads, First-Order IIR, AGC, Filter B

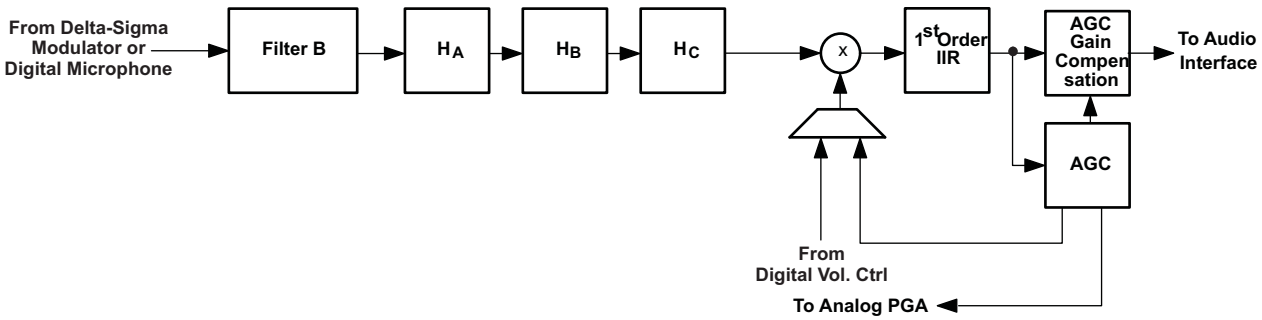


Figure 34. Signal Chain for PRB_R8 and PRB_R11

10.3.12.2.6 20-Tap FIR, First-Order IIR, AGC, Filter B

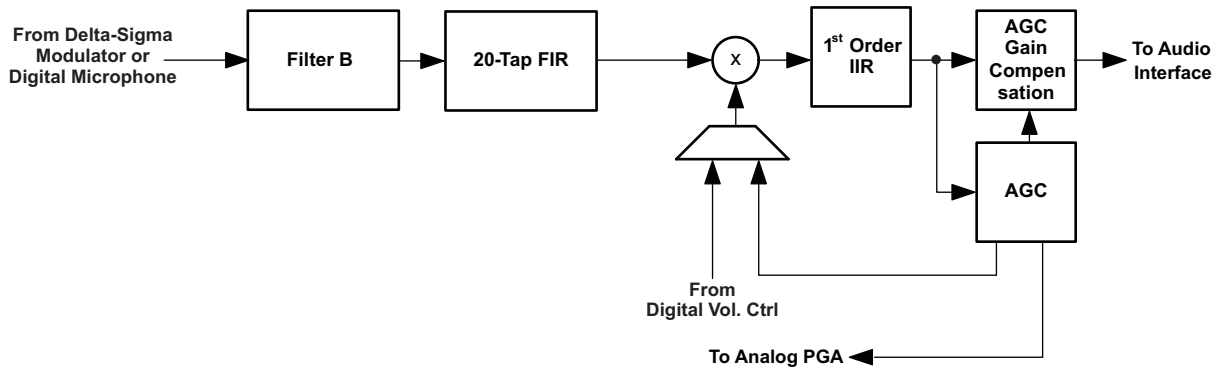


Figure 35. Signal Chain for PRB_R9 and PRB_R12

10.3.12.2.7 First-Order IIR, AGC, Filter C

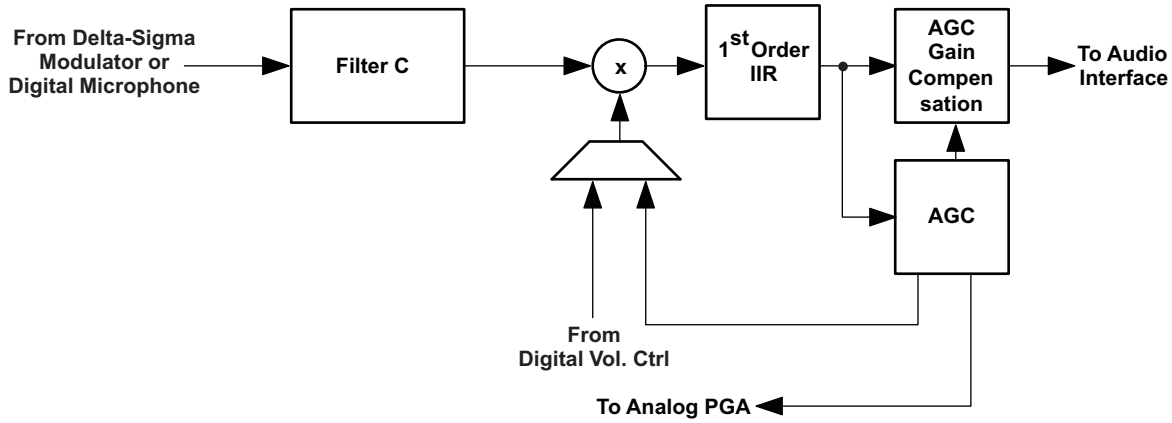


Figure 36. Signal Chain for PRB_R13 and PRB_R16

Copyright © 2008–2015, Texas Instruments Incorporated

10.3.12.2.8 Five Biquads, First-Order IIR, AGC, Filter C

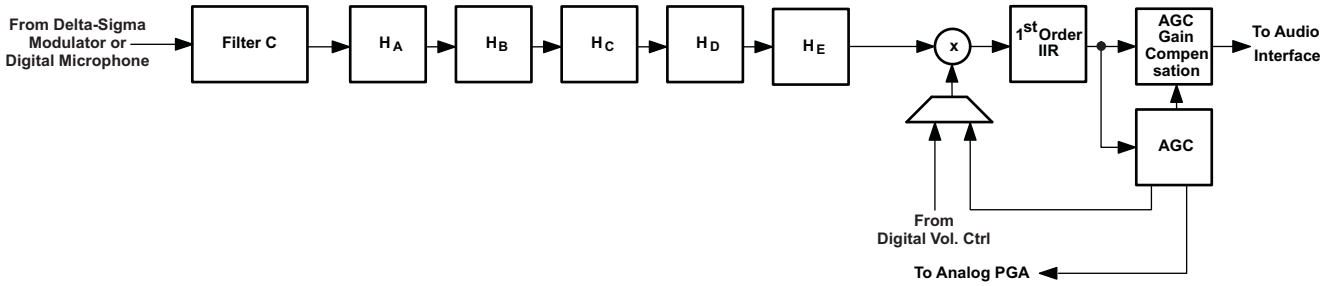


Figure 37. Signal Chain for PRB_R14 and PRB_R17

10.3.12.2.9 25-Tap FIR, First-Order IIR, AGC, Filter C

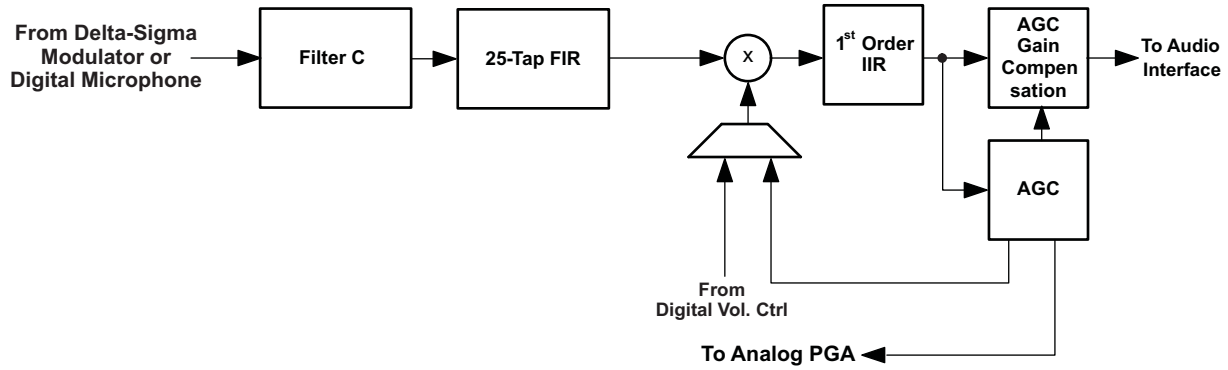


Figure 38. Signal for PRB_R15 and PRB_R18

10.3.12.3 User-Programmable Filters

Depending on the selected processing block, different types and orders of digital filtering are available. A first-order IIR filter is always available, and is useful to filter out possible dc components of the signal efficiently. Up to five biquad sections, or alternatively up to 25-tap FIR filters, are available for specific processing blocks. The coefficients of the available filters are arranged as sequentially indexed coefficients in two banks. If adaptive filtering is chosen, the coefficient banks can be switched in real time.

The coefficients of these filters are each 16-bits wide, in 2s-complement and occupy two consecutive 8-bit registers in the register space. Table 7.

10.3.12.3.1 First-Order IIR Section

The transfer function for the first-order IIR filter is given by Equation 3.

$$H(z) = \frac{N_0 + N_1 z^{-1}}{2^{15} - D_1 z^{-1}} \quad (3)$$

The frequency response for the first-order IIR section with default coefficients is flat at a gain of 0 dB.

Table 7. ADC 1st order IIR Filter Coefficients

FILTER	FILTER COEFFICIENT	ADC COEFFICIENT, LEFT CHANNEL	ADC COEFFICIENT, RIGHT CHANNEL
First-order IIR	N0	C4 (page 4 / register 8 and page 4 / register 9)	C36 (page 4 / register 72 and page 4 / register 73)
	N1	C5 (page 4 / register 10 and page 4 / register 11)	C37 (page 4 / register 74 and page 4 / register 75)
	D1	C6 (page 4 / register 12 and page 4 / register 13)	C38 (page 4 / register 76 and page 4 / register 77)

Copyright © 2008–2015, Texas Instruments Incorporated

10.3.12.3.2 Biquad Section

The transfer function of each of the biquad filters is given by Equation 4.

$$H(z) = \frac{N_0 + 2 \times N_1 z^{-1} + N_2 z^{-2}}{2^{15} - 2 \times D_1 z^{-1} - D_2 z^{-2}} \quad (4)$$

The frequency response for each of the biquad sections with default coefficients is flat at a gain of 0 dB.

Table 8. ADC Biquad Filter Coefficients

FILTER	FILTER COEFFICIENT	ADC COEFFICIENT, LEFT CHANNEL	ADC COEFFICIENT, RIGHT CHANNEL
Biquad A	N0	C7 (page 4 / register 14 and page 4 / register 15)	C39 (page 4 / register 78 and page 4 / register 79)
	N1	C8 (page 4 / register 16 and page 4 / register 17)	C40 (page 4 / register 80 and page 4 / register 81)
	N2	C9 (page 4 / register 18 and page 4 / register 19)	C41 (page 4 / register 82 and page 4 / register 83)
	D1	C10 (page 4 / register 20 and page 4 / register 21)	C42 (page 4 / register 84 and page 4 / register 85)
	D2	C11 (page 4 / register 22 and page 4 / register 23)	C43 (page 4 / register 86 and page 4 / register 87)
Biquad B	N0	C12 (page 4 / register 24 and page 4 / register 25)	C44 (page 4 / register 88 and page 4 / register 89)
	N1	C13 (page 4 / register 26 and page 4 / register 27)	C45 (page 4 / register 90 and page 4 / register 91)
	N2	C14 (page 4 / register 28 and page 4 / register 29)	C46 (page 4 / register 92 and page 4 / register 93)
	D1	C15 (page 4 / register 30 and page 4 / register 31)	C47 (page 4 / register 94 and page 4 / register 95)
	D2	C16 (page 4 / register 32 and page 4 / register 33)	C48 (page 4 / register 96 and page 4 / register 97)
Biquad C	N0	C17 (page 4 / register 34 and page 4 / register 35)	C49 (page 4 / register 98 and page 4 / register 99)
	N1	C18 (page 4 / register 36 and page 4 / register 37)	C50 (page 4 / register 100 and page 4 / register 101)
	N2	C19 (page 4 / register 38 and page 4 / register 39)	C51 (page 4 / register 102 and page 4 / register 103)
	D1	C20 (page 4 / register 40 and page 4 / register 41)	C52 (page 4 / register 104 and page 4 / register 105)
	D2	C21 (page 4 / register 42 and page 4 / register 43)	C53 (page 4 / register 106 and page 4 / register 107)
Biquad D	N0	C22 (page 4 / register 44 and page 4 / register 45)	C54 (page 4 / register 108 and page 4 / register 109)
	N1	C23 (page 4 / register 46 and page 4 / register 47)	C55 (page 4 / register 110 and page 4 / register 111)
	N2	C24 (page 4 / register 48 and page 4 / register 49)	C56 (page 4 / register 112 and page 4 / register113)
	D1	C25 (page 4 / register 50 and page 4 / register 51)	C57 (page 4 / register 114 and page 4 / register 115)
	D2	C26 (page 4 / register 52 and page 4 / register 53)	C58 (page 4 / register 116 and page 4 / register 117)
Biquad E	N0	C27 (page 4 / register 54 and page 4 / register 55)	C59 (page 4 / register 118 and page 4 / register 119)
	N1	C28 (page 4 / register 56 and page 4 / register 57)	C60 (page 4 / register 120 and page 4 / register 121)
	N2	C29 (page 4 / register 58 and page 4 / register 59)	C61 (page 4 / register 122 and page 4 / register 123)
	D1	C30 (page 4 / register 60 and page 4 / register 61)	C62 (page 4 / register 124 and page 4 / register 125)
	D2	C31 (page 4 / register 62 and page 4 / register 63)	C63 (page 4 / register 126 and page 4 / register 127)

10.3.12.3.3 FIR Section

Six of the available ADC processing blocks offer FIR filters for signal processing. PRB_R9 and PRB_R12 feature a 20-tap FIR filter, whereas the processing blocks PRB_R3, PRB_R6, PRB_R15, and PRB_R18 feature a 25-tap FIR filter

$$H(z) = \sum_{n=0}^{M} Fir_n z^{-n}$$

$$M = 24, \text{for PRB_R3, PRB_R6, PRB_R15 and PRB_R18}$$

$$M = 19, \text{for PRB_R9 and PRB_R12} \quad (5)$$

The coefficients of the FIR filters are 16-bit 2s-complement format and correspond to the ADC coefficient space as listed in Table 9. There is no default transfer function for the FIR filter. When the FIR filter is used, all applicable coefficients must be programmed.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 9. ADC FIR Filter Coefficients

FILTER COEFFICIENT	ADC COEFFICIENT LEFT CHANNEL	ADC COEFFICIENT RIGHT CHANNEL
Fir0	C7 (page 4 / register 14 and page 4 / register 15)	C39 (page 4 / register 78 and page 4 / register 79)
Fir1	C8 (page 4 / register 16 and page 4 / register 17)	C40 (page 4 / register 80 and page 4 / register 81)
Fir2	C9 (page 4 / register 18 and page 4 / register 19)	C41 (page 4 / register 82 and page 4 / register 83)
Fir3	C10 (page 4 / register 20 and page 4 / register 21)	C42 (page 4 / register 84 and page 4 / register 85)
Fir4	C11 (page 4 / register 22 and page 4 / register 23)	C43 (page 4 / register 86 and page 4 / register 87)
Fir5	C12 (page 4 / register 24 and page 4 / register 25)	C44 (page 4 / register 88 and page 4 / register 89)
Fir6	C13 (page 4 / register 26 and page 4 / register 27)	C45 (page 4 / register 90 and page 4 / register 91)
Fir7	C14 (page 4 / register 28 and page 4 / register 29)	C46 (page 4 / register 92 and page 4 / register 93)
Fir8	C15 (page 4 / register 30 and page 4 / register 31)	C47 (page 4 / register 94 and page 4 / register 95)
Fir9	C16 (page 4 / register 32 and page 4 / register 33)	C48 (page 4 / register 96 and page 4 / register 97)
Fir10	C17 (page 4 / register 34 and page 4 / register 35)	C49 (page 4 / register 98 and page 4 / register 99)
Fir11	C18 (page 4 / register 36 and page 4 / register 37)	C50 (page 4 / register 100 and page 4 / register 101)
Fir12	C19 (page 4 / register 38 and page 4 / register 39)	C51 (page 4 / register 102 and page 4 / register 103)
Fir13	C20 (page 4 / register 40 and page 4 / register 41)	C52 (page 4 / register 104 and page 4 / register 105)
Fir14	C21 (page 4 / register 42 and page 4 / register 43)	C53 (page 4 / register 106 and page 4 / register 107)
Fir15	C22 (page 4 / register 44 and page 4 / register 45)	C54 (page 4 / register 108 and page 4 / register 109)
Fir16	C23 (page 4 / register 46 and page 4 / register 47)	C55 (page 4 / register 110 and page 4 / register 111)
Fir17	C24 (page 4 / register 48 and page 4 / register 49)	C56 (page 4 / register 112 and page 4 / register113)
Fir18	C25 (page 4 / register 50 and page 4 / register 51)	C57 (page 4 / register 114 and page 4 / register 115)
Fir19	C26 (page 4 / register 52 and page 4 / register 53)	C58 (page 4 / register 116 and page 4 / register 117)
Fir20	C27 (page 4 / register 54 and page 4 / register 55)	C59 (page 4 / register 118 and page 4 / register 119)
Fir21	C28 (page 4 / register 56 and page 4 / register 57)	C60 (page 4 / register 120 and page 4 / register 121)
Fir22	C29 (page 4 / register 58 and page 4 / register 59)	C61 (page 4 / register 122 and page 4 / register 123)
Fir23	C30 (page 4 / register 60 and page 4 / register 61)	C62 (page 4 / register 124 and page 4 / register 125)
Fir24	C31 (page 4 / register 62 and page 4 / register 63)	C63 (page 4 / register 126 and page 4 / register 127)

10.3.12.4 Decimation Filter

The TLV320ADC3001 offers three different types of decimation filters. The integrated digital decimation filter removes high-frequency content and downsamples the audio data from an initial sampling rate of AOSR × f_S to the final output sampling rate of f_S. The decimation filtering is achieved using a higher-order CIC filter followed by linear-phase FIR filters. The decimation filter cannot be chosen by itself, it is implicitly set through the chosen processing block.

The following subsections describe the properties of the available filters A, B, and C.

Copyright © 2008–2015, Texas Instruments Incorporated

10.3.12.4.1 Decimation Filter A

This filter is intended for use at sampling rates up to 48 kHz. When configuring this filter, the oversampling ratio of the ADC can either be 128 or 64. For highest performance, the oversampling ratio must be set to 128. Filter A can also be used for 96 kHz at an AOSR of 64.

Table 10. ADC Decimation Filter A, Specification

PARAMETER	CONDITION	VALUE (TYPICAL)	UNIT
AOSR = 128			
Filter gain pass band	0…0.39 f_S	0.062	dB
Filter gain stop band	0.55…64 f_S	–73	dB
Filter group delay		17/f_S	s
Pass-band ripple, 8 ksps	0…0.39 f_S	0.062	dB
Pass-band ripple, 44.18 ksps	0…0.39 f_S	0.05	dB
Pass-band ripple, 48 ksps	0…0.39 f_S	0.05	dB
AOSR = 64			
Filter gain pass band	0…0.39 f_S	0.062	dB
Filter gain stop band	0.55…32 f_S	–73	dB
Filter group delay		17/f_S	s
Pass-band ripple, 8 ksps	0…0.39 f_S	0.062	dB
Pass-band ripple, 44.18 ksps	0…0.39 f_S	0.05	dB
Pass-band ripple, 48 ksps	0…0.39 f_S	0.05	dB
Pass-band ripple, 96 ksps	0…20 kHz	0.1	dB

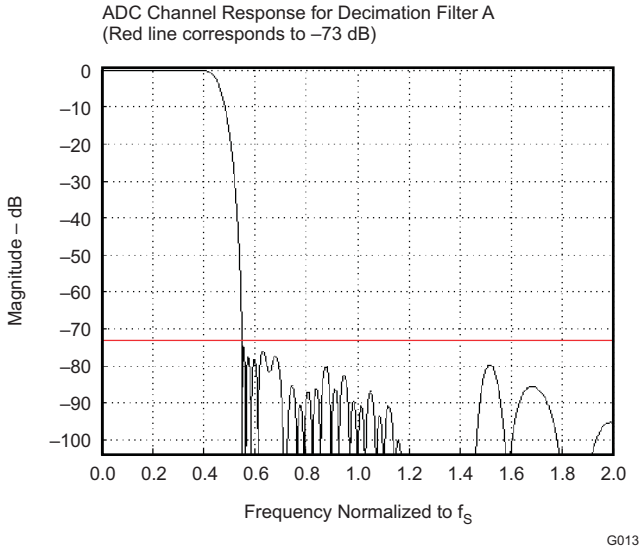


Figure 39. ADC Decimation Filter A, Frequency Response

Copyright © 2008–2015, Texas Instruments Incorporated

10.3.12.4.2 Decimation Filter B

Filter B is intended to support sampling rates up to 96kHz at a oversampling ratio of 64.

Table 11. ADC Decimation Filter B, Specifications

PARAMETER	CONDITION	VALUE (TYPICAL)	UNIT
AOSR = 64			
Filter gain pass band	0…0.39 f_S	±0.077	dB
Filter gain stop band	0.60 f_S…32 f_S	–46	dB
Filter group delay		11/f_S	s
Pass-band ripple, 8 ksps	0…0.39 f_S	0.076	dB
Pass-band ripple, 44.18 ksps	0…0.39 f_S	0.06	dB
Pass-band ripple, 48 ksps	0…0.39 f_S	0.06	dB
Pass-band ripple, 96 ksps	0…20kHz	0.11	dB

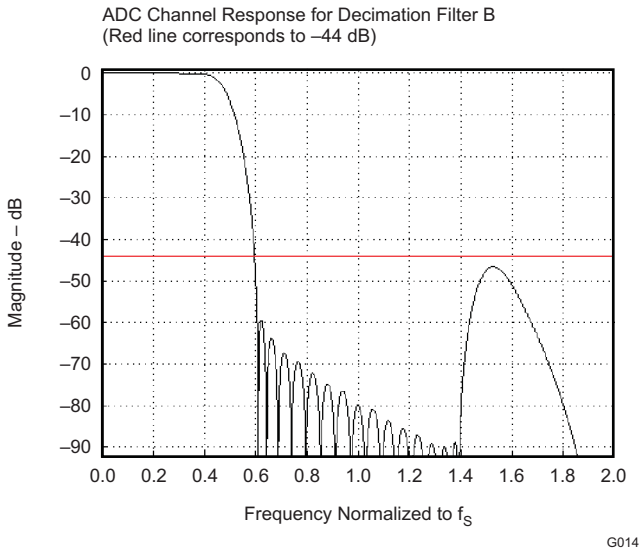


Figure 40. ADC Decimation Filter B, Frequency Response

10.3.12.4.3 Decimation Filter C

Filter type C along with AOSR of 32 is specially designed for 192ksps operation for the ADC. The pass band which extends up to 0.11 × f_S (corresponds to 21 kHz), is suited for audio applications.

Table 12. ADC Decimation Filter C, Specifications

PARAMETER	CONDITION	VALUE (TYPICAL)	UNIT
Filter gain from 0 to 0.11 f_S	0…0.11 f_S	±0.033	dB
Filter gain from 0.28 f_S to 16 f_S	0.28 f_S…16 f_S	–60	dB
Filter group delay		11/ f_S	s
Pass-band ripple, 8 ksps	0…0.11 f_S	0.033	dB
Pass-band ripple, 44.18 ksps	0…0.11 f_S	0.033	dB
Pass-band ripple, 48 ksps	0…0.11 f_S	0.032	dB
Pass-band ripple, 96 ksps	0…0.11 f_S	0.032	dB
Pass-band ripple, 192 ksps	0…20 kHz	0.086	dB

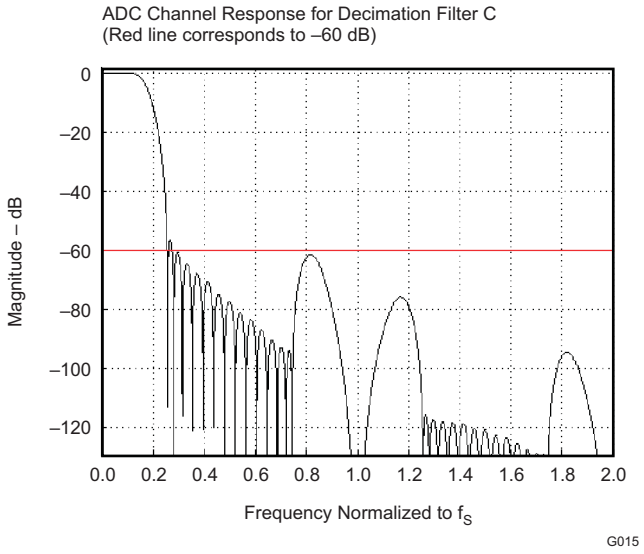


Figure 41. ADC Decimation Filter C, Frequency Response

10.4 Device Functional Modes

10.4.1 Recording Mode

The recording mode is activated once the ADC blocks are enabled. The record path operates from 8 kHz to 48 kHz in single-rate mode and up to 96 kHz in dual-rate mode. It contains programmable input channel configurations supporting single-ended and differential setups. To provide optimal system power management, the stereo recording path can be powered up one channel at a time, to support the case where only mono record capability is required. Digital signal processing blocks can remove audible noise that may be introduced by mechanical coupling. The TLV320ADC3001 includes Automatic Gain Control (AGC) for ADC recording.

10.5 Programming

10.5.1 Digital Interfaces

10.5.1.1 I^2C Control Mode

The TLV320ADC3001 supports the I^2C control protocol using 7-bit addressing and is capable of operating in both standard mode (≤ 100 kHz) and fast mode (≤ 400 kHz). The device address is fixed with the value 0011 000.

Copyright © 2008–2015, Texas Instruments Incorporated

Programming (continued)

I^2C is a two-wire, open-drain interface supporting multiple devices and masters on a single bus. Devices on the I^2C bus only drive the bus lines LOW by connecting them to ground; they never drive the bus lines HIGH. Instead, the bus wires are pulled HIGH by pullup resistors, so the bus wires are HIGH when no device is driving them LOW. This way, two devices cannot conflict; if two devices drive the bus simultaneously, there is no driver contention.

Communication on the I^2C bus always takes place between two devices, one acting as the master and the other acting as the slave. Both masters and slaves can read and write, but slaves can only do so under the direction of a master. Some I^2C devices can act as masters or slaves, but the TLV320ADC3001 can only act as a slave device.

An I^2C bus consists of two lines, SDA and SCL. SDA carries data; SCL provides the clock. All data is transmitted across the I^2C bus in groups of eight bits. To send a bit on the I^2C bus, the SDA line is driven to the appropriate level while SCL is LOW (a LOW on SDA indicates the bit is 0; a HIGH indicates the bit is 1). Once the SDA line has settled, the SCL line is brought HIGH, then LOW. This pulse on SCL clocks the SDA bit into the receiver shift register.

The I^2C bus is bidirectional: the SDA line is used both for transmitting and receiving data. When a master reads from a slave, the slave drives the data line; when a master sends to a slave, the master drives the data line. Under normal circumstances, the master drives the clock line.

Most of the time the bus is idle, no communication is taking place, and both lines are HIGH. When communication is taking place, the bus is active. Only master devices can start a communication. They do this by causing a START condition on the bus. Normally, the data line is only allowed to change state while the clock line is LOW. If the data line changes state while the clock line is HIGH, it is either a START condition or its counterpart, a STOP condition. A START condition is when the clock line is HIGH and the data line goes from HIGH to LOW. A STOP condition is when the clock line is HIGH and the data line goes from LOW to HIGH.

After the master issues a START condition, it sends a byte that indicates the slave device with which it is to communicate. This byte is called the address byte. Each device on an I^2C bus has a unique 7-bit address to which it responds. (Slaves can also have 10-bit addresses; see the I^2C specification for details.) The master sends an address in the address byte, together with a bit that indicates whether it is to read from or write to the slave device.

Every byte transmitted on the I^2C bus, whether it is address or data, is acknowledged with an acknowledge bit. When a master has finished sending a byte (eight data bits) to a slave, it stops driving SDA and waits for the slave to acknowledge the byte. The slave acknowledges the byte by pulling SDA LOW. The master then sends a clock pulse to clock the acknowledge bit. Similarly, when a master has finished reading a byte, it pulls SDA LOW to acknowledge this to the slave. It then sends a clock pulse to clock the bit.

A not-acknowledge is performed by leaving SDA HIGH during an acknowledge cycle. If a device is not present on the bus, and the master attempts to address it, it receives a not-acknowledge because no device is present at that address to pull the line LOW.

When a master has finished communicating with a slave, it may issue a STOP condition. When a STOP condition is issued, the bus becomes idle again. A master may also issue another START condition. When a START condition is issued while the bus is active, it is called a repeated START condition.

The TLV320ADC3001 also responds to and acknowledges a general call, which consists of the master issuing a command with a slave address byte of 00h.

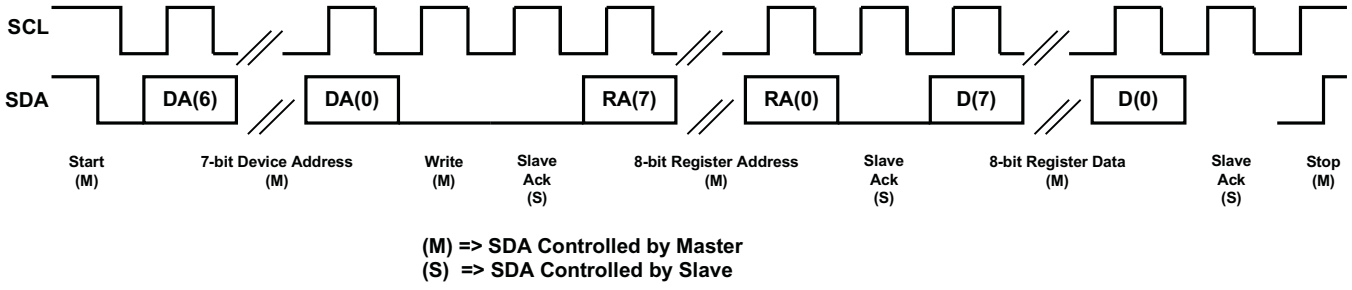


Figure 42. I^2C Write

Programming (continued)

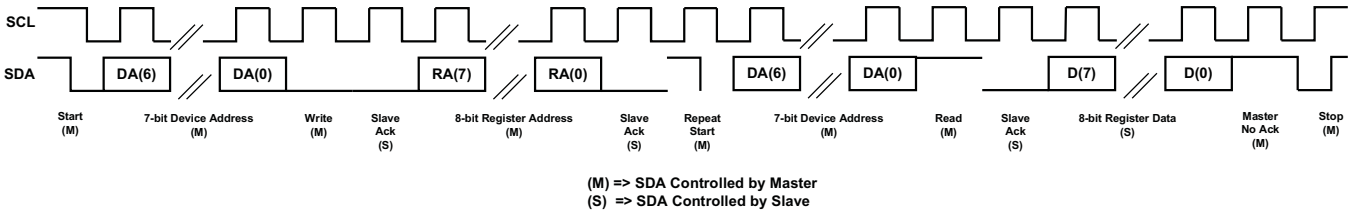


Figure 43. I²C Read

In the case of an I²C register write, if the master does not issue a STOP condition, then the device enters auto-increment mode. So in the next eight clocks, the data on SDA is treated as data for the next incremental register.

Similarly, in the case of an I²C register read, after the device has sent out the 8-bit data from the addressed register, if the master issues an ACKNOWLEDGE, the slave takes over control of SDA bus and transmits for the next eight clocks the data of the next incremental register.

10.6 Register Maps

10.6.1 Control Registers

The control registers for the TLV320ADC3001 are described in detail as follows. All registers are 8 bits in width, with D7 referring to the most-significant bit of each register, and D0 referring to the least-significant bit.

Pages 0, 1, 4, 5, and 32–47 are available. All other pages are reserved. Do not read from or write to reserved pages.

Copyright © 2008–2015, Texas Instruments Incorporated

Register Maps (continued)

Table 13. Page / Register Map

REGISTER NO.	REGISTER NAME
PAGE 0: (Clock Multipliers and Dividers, Serial Interfaces, Flags, Interrupts, and Programming of GPIOs)	
0	Page control register
1	S/W RESET
2	Reserved
3	Reserved
4	Clock-gen muxing
5	PLL P and R-VAL
6	PLL J-VAL
7	PLL D-VAL MSB
8	PLL D-VAL LSB
9–17	Reserved
18	ADC NADC clock divider
19	ADC MADC clock divider
20	ADC AOSR
21	ADC IADC
22	ADC miniDSP engine decimation
23 and 24	Reserved
25	CLKOUT MUX
26	CLKOUT M divider
27	ADC interface control 1
28	DATA slot offset programmability 1 (Ch_Offset_1)
29	ADC interface control 2
30	BCLK N divider
31–33	Reserved
34	I^2S sync
35	Reserved
36	ADC flag register
37	Data slot offset programmability 2 (Ch_Offset_2)
38	I^2S TDM control register
39–41	Reserved
42	Interrupt flags (overflow)
43	Interrupt flags (overflow)
44	Reserved
45	Interrupt flags-ADC
46	Reserved
47	Interrupt flags-ADC
48–52	Reserved
53	DOUT (Out pin) Control
54–56	Reserved
57	ADC sync control 1
58	ADC sync control 2
59	ADC CIC filter gain control
60	Reserved
61	ADC processing block / miniDSP selection
62	Programmable instruction-mode control bits
63–80	Reserved

Copyright © 2008–2015, Texas Instruments Incorporated

Register Maps (continued)

Table 13. Page / Register Map (continued)

REGISTER NO.	REGISTER NAME
81	ADC digital
82	ADC fine volume control
83	Left ADC volume control
84	Right ADC volume control
85	ADC phase compensation
86	Left AGC control 1
87	Left AGC control 2
88	Left AGC maximum gain
89	Left AGC attack time
90	Left AGC decay time
91	Left AGC noise debounce
92	Left AGC signal debounce
93	Left AGC gain
94	Right AGC control 1
95	Right AGC control 2
96	Right AGC maximum gain
97	Right AGC attack time
98	Right AGC decay time
99	Right AGC noise debounce
100	Right AGC signal debounce
101	Right AGC gain
102–127	Reserved
PAGE 1: (ADC ROUTING, PGA, POWER CONTROLS, AND MISC LOGIC-RELATED PROGRAMMABILITIES)	
0	Page control register
1–25	Reserved
26	Dither control
27–50	Reserved
51	MICBIAS control
52	Left ADC input selection for left PGA
53	Reserved
54	Left ADC input selection for left PGA
55	Right ADC input selection for right PGA
56	Reserved
57	Right ADC input selection for right PGA
58	Reserved
59	Left analog PGA setting
60	Right analog PGA setting
61	ADC low-current modes
62	ADC analog PGA flags
63–127	Reserved
PAGE 2: Reserved. Do not read from or write to this page.	
PAGE 3: Reserved. Do not read from or write to this page.	

Copyright © 2008–2015, Texas Instruments Incorporated

Register Maps (continued)

Table 13. Page / Register Map (continued)

REGISTER NO.	REGISTER NAME
ADC Digital Filter RAM and Instruction Pages:	
PAGE 4: ADC Programmable Coefficients RAM (1:63)	
PAGE 5: ADC Programmable Coefficients RAM (65:127)	
PAGE 6–PAGE 31: Reserved. Do not read from or write to these pages.	
PAGE 32–PAGE 47: ADC Programmable Instruction RAM (0:511)	
Page 32 Instruction Inst(0:31)	
Page 33 Instruction Inst(32:63)	
Page 34 Instruction Inst(64:95)	
...	
Page 47 Instruction Inst(480:511)	
PAGE 48–PAGE 255: Reserved. Do not read from or write to these pages.	

10.6.2 Control Registers, Page 0: Clock Multipliers and Dividers, Serial Interfaces, Flags, Interrupts and Programming of GPIOs

Table 14. Page 0 / Register 0: Page Control Register(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Page 0 selected 0000 0001: Page 1 selected ... 1111 1110: Page 254 selected (reserved) 1111 1111: Page 255 selected (reserved)

(1) Valid pages are 0, 1, 4, 5, 32–47. All other pages are reserved (do not access).

Table 15. Page 0 / Register 1: Software Reset

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D1	R	0000 000	Reserved. Write only zeros to these bits.
D0	W	0	0: Don't care 1: Self-clearing software reset for control register

Table 16. Page 0 / Register 2: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	0000 0000	Reserved. Do not write any value other than reset value.

Table 17. Page 0 / Register 3: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 18. Page 0 / Register 4: Clock-Gen Multiplexing(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R	0000	Reserved. Do not write any value other than reset value.
D3–D2	R/W	00	00: PLL_CLKIN = MCLK (device pin) 01: PLL_CLKIN = BCLK (device pin) 10: Reserved. Do not use. 11: PLL_CLKIN = logic level 0
D1–D0	R/W	00	00: CODEC_CLKIN = MCLK (device pin) 01: CODEC_CLKIN = BCLK (device pin) 10: Reserved. Do not use. 11: CODEC_CLKIN = PLL_CLK (generated on-chip)

(1) Refer to Figure 27 for more details on clock generation multiplexing and dividers.

Table 19. Page 0 / Register 5: PLL P and R-VAL

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: PLL is powered down. 1: PLL is powered up.
D6–D4	R/W	001	000: PLL divider P = 8 001: PLL divider P = 1 010: PLL divider P = 2 ... 110: PLL divider P = 6 111: PLL divider P = 7
D3–D0	R/W	0001	0000: PLL multiplier R = 16 0001: PLL multiplier R = 1 0010: PLL multiplier R = 2 ... 1110: PLL multiplier R = 14 1111: PLL multiplier R = 15

Table 20. Page 0 / Register 6: PLL J-VAL

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D6	R/W	00	Reserved. Write only zeros to these bits.
D5–D0	R/W	00 0100	00 0000: Don't use (reserved) 00 0001: PLL multiplier J = 1 00 0010: PLL multiplier J = 2 00 0011: PLL multiplier J = 3 00 0100: PLL multiplier J = 4 (default) ... 11 1110: PLL multiplier J = 62 11 1111: PLL multiplier J = 63

Table 21. Page 0 / Register 7: PLL D-VAL MSB(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D6	R/W	00	Reserved. Write only zeros to these bits.
D5–D0	R/W	00 0000	PLL fractional multiplier bits D13–D8

(1) Page 0 / register 7 is updated when page 0 / register 8 is written immediately after page 0 / register 7 is written.

Table 22. Page 0 / Register 8: PLL D-VAL LSB(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	PLL fractional multiplier bits D7–D0

(1) Page 0 / register 8 must be written immediately after writing to page 0 / register 7.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 23. Page 0 / Register 9 Through Page 0 / Register 17: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Table 24. Page 0 / Register 18: ADC NADC Clock Divider

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	NADC Clock-Divider Power Control: 0: NADC clock divider is powered down 1: NADC clock divider is powered up
D6–D0	R/W	000 0001	NADC Value: 000 0000: NADC clock divider = 128 000 0001: NADC clock divider = 1 000 0010: NADC clock divider = 2 ... 111 1110: NADC clock divider = 126 111 1111: NADC clock divider = 127

Table 25. Page 0 / Register 19: ADC MADC Clock Divider

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: ADC MADC clock divider is powered down 1: ADC MADC clock divider is powered up
D6–D0	R/W	000 0001	000 0000: MADC clock divider = 128 000 0001: MADC clock divider = 1 000 0010: MADC clock divider = 2 ... 111 1110: MADC clock divider = 126 111 1111: MADC clock divider = 127

Table 26. Page 0 / Register 20: ADC AOSR(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	1000 0000	ADC Oversampling Value (AOSR): 0000 0000: AOSR = 256 0000 0001: AOSR = 1 0000 0010: AOSR = 2 ... 1111 1110: AOSR = 254 1111 1111: AOSR = 255

(1) AOSR must be an integral multiple of the ADC decimation factor.

Table 27. Page 0 / Register 21: ADC IADC(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	1000 0000	Number of instructions for ADC miniDSP (IADC): 0000 0000: Reserved. Do not use. 0000 0001: IADC = 2 0000 0010: IADC = 4 ... 1111 1110: IADC = 508 1111 1111: IADC = 510

(1) IADC must be an integral multiple of the ADC decimation factor.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 28. Page 0 / Register 22: ADC miniDSP Engine Decimation

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R	0000	Reserved. Do not write any value other than reset value.
D3–D0	R/W	0100	0000: Decimation ratio in ADC miniDSP engine = 16 0001: Decimation ratio in ADC miniDSP engine = 1 0010: Decimation ratio in ADC miniDSP engine = 2 ... 1101: Decimation ratio in ADC miniDSP engine = 13 1110: Decimation ratio in ADC miniDSP engine = 14 1111: Decimation ratio in ADC miniDSP engine = 15

Table 29. Page 0 / Register 23 Through Page 0 / Register 24: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Table 30. Page 0 / Register 25: CLKOUT MUX

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D3	R	0000 0	Reserved. Do not write any value other than reset value.
D2-D0	R/W	000	000: CDIV_CLKIN = MCLK (device pin) 001: CDIV_CLKIN = BCLK (device pin) 010: Reserved. Do not use. 011: CDIV_CLKIN = PLL_CLK (generated on-chip) 100: Reserved. Do not use. 101: Reserved. Do not use. 110: CDIV_CLKIN = ADC_CLK (generated on-chip) 111: CDIV_CLKIN = ADC_MOD_CLK (generated on-chip)

Table 31. Page 0 / Register 26: CLKOUT M Divider

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: CLKOUT M divider is powered down. 1: CLKOUT M divider is powered up.
D6–D0	R/W	000 0001	000 0000: CLKOUT divider M = 128 000 0001: CLKOUT divider M = 1 000 0010: CLKOUT divider M = 2 ... 111 1110: CLKOUT divider M = 126 111 1111: CLKOUT divider M = 127

Table 32. Page 0 / Register 27: ADC Audio Interface Control 1

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D6	R/W	00	00: ADC interface = I^2S 01: ADC interface = DSP 10: ADC interface = RJF 11: ADC interface = LJF
D5–D4	R/W	00	00: ADC interface word length = 16 bits 01: ADC interface word length = 20 bits 10: ADC interface word length = 24 bits 11: ADC interface word length = 32 bits
D3	R/W	0	0: BCLK is input. 1: BCLK is output.
D2	R/W	0	0: WCLK is input. 1: WCLK is output.
D1	R	0	Reserved. Do not write any value other than reset value.
D0	R/W	0	0: Hi-Z operation of DOUT: disabled 1: Hi-Z operation of DOUT: enabled

Copyright © 2008–2015, Texas Instruments Incorporated

Table 33. Page 0 / Register 28: Data Slot Offset Programmability 1 (Ch_Offset_1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Offset = 0 BCLKs. Offset is measured with respect to WCLK rising edge in DSP mode.(1) 0000 0001: Offset = 1 BCLKs 0000 0010: Offset = 2 BCLKs ... 1111 1110: Offset = 254 BCLKs 1111 1111: Offset = 255 BCLKs

(1) Usage controlled by page 0 / register 38, bit D0

Table 34. Page 0 / Register 29: ADC Interface Control 2

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R/W	0000	Reserved. Do not write any value other than reset value.
D3	R/W	0	0: BCLK is not inverted (valid for both primary and secondary BCLK). 1: BCLK is inverted (valid for both primary and secondary BCLK).
D2	R/W	0	0: BCLK and WCLK active even with codec powered down: disabled (valid for both primary and secondary BCLK) 1: BCLK and WCLK active even with codec powered down: enabled (valid for both primary and secondary BCLK)
D1–D0	R/W	10	00: Reserved. Do not use. 01: Reserved. Do not use. 10: BDIV_CLKIN = ADC_CLK (generated on-chip) 11: BDIV_CLKIN = ADC_MOD_CLK (generated on-chip)

Table 35. Page 0 / Register 30: BCLK N Divider

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: BCLK N divider is powered down. 1: BCLK N divider is powered up.
D6–D0	R/W	000 0001	000 0000: CLKOUT divider N = 128 000 0001: CLKOUT divider N = 1 000 0010: CLKOUT divider N = 2 ... 111 1110: CLKOUT divider N = 126 111 1111: CLKOUT divider N = 127

Table 36. Page 0 / Register 31 Through Page 0 / Register 33: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 37. Page 0 / Register 34: I^2S Sync

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: Internal logic is enabled to detect the I^2C hang and react accordingly. 1: Internal logic is disabled to detect the I^2C hang.
D6	R	0	0: I^2C hang is not detected. 1: I^2C hang is detected. D6 bit is cleared to "0" only by reading this register
D5	R/W	0	0: I^2C general-call address is ignored. 1: Device accepts I^2C general-call address.
D4–D2	R	000	Reserved. Do not write any value other than reset value.
D1	R/W	0	0: Re-sync logic is disabled for ADC. 1: Re-sync stereo ADC with codec interface if the group delay changed by more than ±ADC_f_S/4.
D0	R/W	0	0: Re-sync is done without soft-muting the channel for ADC. 1: Re-sync is done by internally soft-muting the channel for ADC.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 38. Page 0 / Register 35: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 39. Page 0 / Register 36: ADC Flag Register

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R	0	0: Left ADC PGA, applied gain ≠ programmed gain 1: Left ADC PGA, applied gain = programmed gain
D6	R	0	0: Left ADC powered down 1: Left ADC powered up
D5(1)	R	0	0: Left AGC not saturated 1: Left AGC applied gain = maximum applicable gain by left AGC
D4	R	0	Reserved. Do not write any value other than reset value.
D3	R	0	0: Right ADC PGA, applied gain ≠ programmed gain 1: Right ADC PGA, applied gain = programmed gain
D2	R	0	0: Right ADC powered down 1: Right ADC powered up
D1(1)	R	0	0: Right AGC not saturated 1: Right AGC applied gain = maximum applicable gain by right AGC
D0	R	0	Reserved. Do not write any value other than reset value.

(1) Sticky flag bits. These are read-only bits. They are automatically cleared once they are read and are set only if the source trigger occurs again.

Table 40. Page 0 / Register 37: Data Slot Offset Programmability 2 (Ch_Offset_2)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Offset = 0 BCLKs. Offset is measured with respect to the end of the first channel(1) 0000 0001: Offset = 1 BCLK 0000 0010: Offset = 2 BCLKs ... 1111 1110: Offset = 254 BCLKs 1111 1111: Offset = 255 BCLKs

(1) Usage controlled by page 0 / register 38, bit D0, time_slot_mode enable

Table 41. Page 0 / Register 38: I^2S TDM Control Register

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D5	R	000	Reserved. Do not write any value other than reset value.
D4	R/W	0	0: Channel swap disabled 1: Channel swap enabled
D3–D2	R/W	00	00: Both left and right channels enabled 01: Left channel enabled 10: Right channel enabled 11: Both left and right channels disabled
D1	R/W	1	0: early_3-state disabled 1: early_3-state enabled
D0	R/W	0	0: time_slot_mode disabled – both channel offsets controlled by Ch_Offset_1 (page 0 / register 28) 1: time_slot_mode enabled – channel-1 offset controlled by Ch_Offset_1 (page 0 / register 28) and channel-2 offset controlled by Ch_Offset_2 (page 0 / register 37)

Table 42. Page 0 / Register 39 Through Page 0 / Register 41: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 43. Page 0 / Register 42: Interrupt Sticky Flags (Overflow)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R	0000	Reserved
D3(1)	R	0	Left ADC Overflow Flag 0: No overflow in left ADC 1: Overflow has occurred in left ADC since last read of this register.
D2(1)	R	0	Right ADC Overflow Flag 0: No overflow in right ADC 1: Overflow has occurred in right ADC since last read of this register.
D1(1)	R	0	ADC Barrel-Shifter Output-Overflow Flag 0: No overflow in ADC barrel-shifter output 1: Overflow has occurred in ADC barrel-shifter output since last read of this register.
D0	R	0	Reserved

(1) Sticky flag bits. These are read-only bits. They are automatically cleared once they are read and are set only if the source trigger occurs again.

Table 44. Page 0 / Register 43: Interrupt Flags (Overflow)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R	0000	Reserved
D3	R	0	Left ADC Overflow Flag 0: No overflow in left ADC 1: Overflow has occurred in left ADC.
D2	R	0	Right ADC Overflow Flag 0: No overflow in right ADC 1: Overflow has occurred in right ADC.
D1	R	0	ADC Barrel-Shifter Output-Overflow Flag 0: No overflow in ADC barrel-shifter output 1: Overflow in ADC barrel-shifter output
D0	R	0	Reserved

Table 45. Page 0 / Register 44: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 46. Page 0 / Register 45: Interrupt Flags—ADC

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION(1)
D7	R	0	Reserved
D6	R	0	Left AGC Noise Threshold Flag: 0: Left ADC signal power ≥ noise threshold for left AGC 1: Left ADC signal power < noise threshold for left AGC
D5	R	0	Right AGC Noise Threshold Flag: 0: Right ADC signal power ≥ noise threshold for right AGC 1: Right ADC signal power < noise threshold for right AGC
D4	R	0	ADC miniDSP engine standard interrupt-port output
D3	R	0	ADC miniDSP engine auxilliary interrupt-port output
D2–D0	R	000	Reserved

(1) Sticky flag bits. These are read-only bits. They are automatically cleared once they are read and are set only if the source trigger occurs again.

Table 47. Page 0 / Register 46: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 48. Page 0 / Register 47: Interrupt Flags—ADC

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R	0	Reserved
D6	R	0	0: Left ADC signal power ≥ noise threshold for left AGC 1: Left ADC signal power < noise threshold for left AGC
D5	R	0	0: Right ADC signal power ≥ noise threshold for right AGC 1: Right ADC signal power < noise threshold for right AGC
D4	R	0	ADC miniDSP engine standard interrupt-port output. This bit indicates the instantaneous value of the interrupt port at the time of reading the register.
D3	R	0	ADC miniDSP engine auxilliary interrupt-port output. This bit indicates the instantaneous value of the interrupt port at the time of reading the register.
D2–D0	R	000	Reserved

Table 49. Page 0 / Register 48 Through Page 0 / Register 52: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Table 50. Page 0 / Register 53: DOUT (Out Pin) Control

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D5	R	000	Reserved. Do not write any value other than reset value.
D4	R/W	1	0: DOUT bus keeper enabled 1: DOUT bus keeper disabled
D3–D1	R/W	001	000: DOUT disabled (output buffer powered down) 001 DOUT = primary DOUT output for codec interface 010: DOUT = general-purpose output 011: DOUT = CLKOUT output 100: DOUT = INT1 output 101: DOUT = INT2 output 110: DOUT = secondary BCLK output for codec interface 111: DOUT = secondary WCLK output for codec interface
D0	R/W	0	0: DOUT value = 0 when bits D3–D1 are programmed to 010 (general-purpose output) 1: DOUT value = 1 when bits D3–D1 are programmed to 010 (general-purpose output)

Table 51. Page 0 / Register 54 Through Page 0 / Register 56: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 52. Page 0 / Register 57: ADC Sync Control 1

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: Default synchronization 1: Custom synchronization
D6–D0	R/W	000 0000	000 0000: Custom synchronization window size = 0 instructions 000 0001: Custom synchronization window size = 2 instructions (±1 instruction) 000 0010: Custom synchronization window size = 4 instructions (±2 instructions) ... 111 1111: Custom synchronization window size = 254 instructions (±127 instructions)

Copyright © 2008–2015, Texas Instruments Incorporated

Table 53. Page 0 / Register 58: ADC Sync Control 2

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Custom synchronization target = instruction 0 0000 0001: Custom synchronization target = instruction 2 0000 0010: Custom synchronization target = instruction 4 ... 1111 1111: Custom synchronization target = instruction 510

Table 54. Page 0 / Register 59: ADC CIC Filter Gain Control

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R/W	0100	Left CIC filter gain(1)
D3–D0	R/W	0100	Right CIC filter gain(1)

(1) For proper operation, CIC gain must be ≤ 1.
If AOSR {page 0 / register 20} = 64 and (1 ≤ filter mode {page 0 / register 61} ≤ 6), then the reset value of 4 results in CIC gain = 1.
Otherwise, the CIC gain = (AOSR/(64 × miniDSP engine decimation))^4 × 2 $^{(CIC\ filter\ gain\ control)}$ for 0 ≤ CIC filter gain control ≤ 12,
and if CIC filter gain control = 15, CIC gain is automatically set such that for 7 ≤ (AOSR/miniDSP engine decimation) ≤ 64,
0.5 < CIC gain ≤ 1.

Table 55. Page 0 / Register 60: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	Reserved. Do not write to this register.

Table 56. Page 0 / Register 61: ADC Processing Block / miniDSP Selection

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D5		000	Reserved. Do not write any value other than reset value.
D4–D0		0 0001	0 0000: ADC miniDSP programmable instruction mode enabled. 0 0001: Select ADC signal-processing block PRB_R1 0 0010: Select ADC signal-processing block PRB_R2 0 0011: Select ADC signal-processing block PRB_R3 0 0100: Select ADC signal-processing block PRB_R4 0 0101: Select ADC signal-processing block PRB_R5 0 0110: Select ADC signal-processing block PRB_R6 0 0111: Select ADC signal-processing block PRB_R7 0 1000: Select ADC signal-processing block PRB_R8 0 1001: Select ADC signal-processing block PRB_R9 0 1010: Select ADC signal-processing block PRB_R10 0 1011: Select ADC signal-processing block PRB_R11 0 1100: Select ADC signal-processing block PRB_R12 0 1101: Select ADC signal-processing block PRB_R13 0 1110: Select ADC signal-processing block PRB_R14 0 1111: Select ADC signal-processing block PRB_R15 1 0000: Select ADC signal-processing block PRB_R16 1 0001: Select ADC signal-processing block PRB_R17 1 0010: Select ADC signal-processing block PRB_R18 1 0011–1 1111: Reserved. Do not use.

Table 57. Page 0 / Register 62: Programmable Instruction-Mode Control Bits

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R	0	Reserved. Do not write any value other than reset value.
D6	R/W	0	ADC miniDSP engine auxiliary control bit A, which can be used for conditional instructions like JMP
D5	R/W	0	ADC miniDSP engine auxiliary control bit B, which can be used for conditional instructions like JMP
D4	R/W	0	0: ADC instruction-counter reset at the start of the new frame is enabled. 1: ADC instruction-counter reset at the start of the new frame is disabled.
D3–D0	R	0000	Reserved. Do not write any value other than reset value.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 58. Page 0 / Register 63 Through Page 0 / Register 80: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Table 59. Page 0 / Register 81: ADC Digital

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: Left-channel ADC is powered down. 1: Left-channel ADC is powered up.
D6	R/W	0	0: Right-channel ADC is powered down. 1: Right-channel ADC is powered up.
D5–D2	R/W	0000	Reserved. Do not write any value other than reset.
D1–D0	R/W	00	00: ADC channel volume control soft-stepping is enabled for one step per sample period. 01: ADC channel volume control soft-stepping is enabled for one step per two sample periods. 10: ADC channel volume control soft-stepping is disabled. 11: Reserved. Do not use.

Table 60. Page 0 / Register 82: ADC Fine Volume Control

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	1	0: Left ADC channel not muted 1: Left ADC channel muted
D6–D4	R/W	000	000: Left ADC channel fine gain = 0 dB 001: Left ADC channel fine gain = –0.1 dB 010: Left ADC channel fine gain = –0.2 dB 011: Left ADC channel fine gain = –0.3 dB 100: Left ADC channel fine gain = –0.4 dB 101–111: Reserved. Do not use.
D3	R/W	1	0: Right ADC channel not muted 1: Right ADC channel muted
D2–D0	R/W	000	000: Left ADC channel fine gain = 0 dB 001: Left ADC channel fine gain = –0.1 dB 010: Left ADC channel fine gain = –0.2 dB 011: Left ADC channel fine gain = –0.3 dB 100: Left ADC channel fine gain = –0.4 dB 101–111: Reserved. Do not use.

Table 61. Page 0 / Register 83: Left ADC Volume Control

BIT	READ/ WRITE	RESET VALUE(1)	DESCRIPTION
D7	R	0	Reserved. Do not write any value other than reset value.
D6–D0	R/W	000 0000	100 0000–110 1000: Left ADC channel volume = 0 dB 110 1000: Left ADC channel volume = –12 dB 110 1001: Left ADC channel volume = –11.5 dB 110 1010: Left ADC channel volume = –11 dB ... 111 1111: Left ADC channel volume = –0.5 dB 000 0000: Left ADC channel volume = –0 dB 000 0001: Left ADC channel volume = 0.5 dB ... 010 0110: Left ADC channel volume = 19 dB 010 0111: Left ADC channel volume = 19.5 dB 010 1000: Left ADC channel volume = 20 dB 010 1001–011 1111: Reserved. Do not use.

(1) Values in 2s-complement decimal format

Copyright © 2008–2015, Texas Instruments Incorporated

Table 62. Page 0 / Register 84: Right ADC Volume Control

BIT	READ/ WRITE	RESET VALUE(1)	DESCRIPTION
D7	R	0	Reserved. Do not write any value other than reset value.
D6–D0	R/W	000 0000	100 0000–110 1000: Right ADC channel volume = 0 dB 110 1000: Rght ADC channel volume = –12 dB 110 1001: Right ADC channel volume = –11.5 dB 110 1010: Rght ADC channel volume = –11 dB ... 111 1111: Right ADC channel volume = –0.5 dB 000 0000: Right ADC channel volume = –0.0 dB 000 0001: Right ADC channel volume = 0.5 dB ... 010 0110: Right ADC channel volume = 19 dB 010 0111: Right ADC channel volume = 19.5 dB 010 1000: Right ADC channel volume = 20 dB 010 1001–011 1111 : Reserved. Do not use.

(1) Values in 2s-complement decimal format

Table 63. Page 0 / Register 85: Left ADC Phase Compensation

BIT	READ/ WRITE	RESET VALUE(1)	DESCRIPTION
D7–D0	R/W	0000 0000	1000 0000: Left ADC has a phase shift of –128 ADC_MOD_CLK cycles with respect to right ADC. 1000 0001: Left ADC has a phase shift of –127 ADC_MOD_CLK cycles with respect to right ADC. ... 1111 1110: Left ADC has a phase shift of –2 ADC_MOD_CLK cycles with respect to right ADC. 1111 1111: Left ADC has a phase shift of –1 ADC_MOD_CLK cycles with respect to right ADC. 0000 0000: No phase shift between stereo ADC channels 0000 0001: Left ADC has a phase shift of 1 ADC_MOD_CLK cycles with respect to right ADC. 0000 0010: Left ADC has a phase shift of 2 ADC_MOD_CLK cycles with respect to right ADC. ... 0111 1110: Left ADC has a phase shift of 126 ADC_MOD_CLK cycles with respect to right ADC. 0111 1111: Left ADC has a phase shift of 127 ADC_MOD_CLK cycles with respect to right ADC.

(1) Values in 2s-complement decimal format

Table 64. Page 0 / Register 86: Left AGC Control 1

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: Left AGC disabled 1: Left AGC enabled
D6–D4	R/W	000	000: Left AGC target level = –5.5 dB 001: Left AGC target level = –8 dB 010: Left AGC target level = –10 dB 011: Left AGC target level = –12 dB 100: Left AGC target level = –14 dB 101: Left AGC target level = –17 dB 110: Left AGC target level = –20 dB 111: Left AGC target level = –24 dB
D3–D0	R	0000	Reserved. Do not write any value other than reset value.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 65. Page 0 / Register 87: Left AGC Control 2

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D6	R/W	00	00: Left AGC hysteresis setting of 1 dB 01: Left AGC hysteresis setting of 2 dB 10: Left AGC hysteresis setting of 4 dB 11: Left AGC hysteresis disabled
D5–D1	R/W	00 000	00 000: Left AGC noise/silence detection is disabled. 00 001: Left AGC noise threshold = –30 dB 00 010: Left AGC noise threshold = –32 dB 00 011: Left AGC noise threshold = –34 dB ... 11 101: Left AGC noise threshold = –86 dB 11 110: Left AGC noise threshold = –88 dB 11 111: Left AGC noise threshold = –90 dB
D0	R/W	0	0: Disable clip stepping for AGC 1: Enable clip stepping for AGC

Table 66. Page 0 / Register 88: Left AGC Maximum Gain

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R	0	Reserved. Do not write any value other than reset value.
D6–D0	R/W	111 1111	000 0000: Left AGC maximum gain = 0 dB 000 0001: Left AGC maximum gain = 0.5 dB 000 0010: Left AGC maximum gain = 1 dB ... 101 0000: Left AGC maximum gain = 40 dB 101 0001–111 1111: Reserved. Do not use.

Table 67. Page 0 / Register 89: Left AGC Attack Time

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D3	R/W	0000 0	0000 0: Left AGC attack time = 1 × (32/f_S) 0000 1: Left AGC attack time = 3 × (32/f_S) 0001 0: Left AGC attack time = 5 × (32/f_S) 0001 1: Left AGC attack time = 7 × (32/f_S) 0010 0: Left AGC attack time = 9 × (32/f_S) ... 1111 0: Left AGC attack time = 61 × (32/f_S) 1111 1: Left AGC attack time = 63 × (32/f_S)
D2–D0	R/W	000	000: Multiply factor for the programmed left AGC attack time = 1 001: Multiply factor for the programmed left AGC attack time = 2 010: Multiply factor for the programmed left AGC attack time = 4 ... 111: Multiply factor for the programmed left AGC attack time = 128

Table 68. Page 0 / Register 90: Left AGC Decay Time

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D3	R/W	0000 0	0000 0: Left AGC decay time = 1 × (512/f_S) 0000 1: Left AGC decay time = 3 × (512/f_S) 0001 0: Left AGC decay time = 5 × (512/f_S) 0001 1: Left AGC decay time = 7 × (512/f_S) 0010 0: Left AGC decay time = 9 × (512/f_S) ... 1111 0: Left AGC decay time = 61 × (512/f_S) 1111 1: Left AGC decay time = 63 × (512/f_S)
D2–D0	R/W	000	000: Multiply factor for the programmed left AGC decay time = 1 001: Multiply factor for the programmed left AGC decay time = 2 010: Multiply factor for the programmed left AGC decay time = 4 ... 111: Multiply factor for the programmed left AGC decay time = 128

Copyright © 2008–2015, Texas Instruments Incorporated

Table 69. Page 0 / Register 91: Left AGC Noise Debounce

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D5	R	000	Reserved. Do not write any value other than reset value.
D4–D0	R/W	0 0000	0 0000: Left AGC noise debounce = 0/f_S 0 0001: Left AGC noise debounce = 4/f_S 0 0010: Left AGC noise debounce = 8/f_S 0 0011: Left AGC noise debounce = 16/f_S 0 0100: Left AGC noise debounce = 32/f_S 0 0101: Left AGC noise debounce = 64/f_S 0 0110: Left AGC noise debounce = 128/f_S 0 0111: Left AGC noise debounce = 256/f_S 0 1000: Left AGC noise debounce = 512/f_S 0 1001: Left AGC noise debounce = 1024/f_S 0 1010: Left AGC noise debounce = 2048/f_S 0 1011: Left AGC noise debounce = 4096/f_S 0 1100: Left AGC noise debounce = 2 × 4096/f_S 0 1101: Left AGC noise debounce = 3 × 4096/f_S 0 1110: Left AGC noise debounce = 4 × 4096/f_S ... 1 1110: Left AGC noise debounce = 20 × 4096/f_S 1 1111: Left AGC noise debounce = 21 × 4096/f_S

Table 70. Page 0 / Register 92: Left AGC Signal Debounce

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R	0000	Reserved. Do not write any value other than reset value.
D3–D0	R/W	0000	0000: Left AGC signal debounce = 0/f_S 0001: Left AGC signal debounce = 4/f_S 0010: Left AGC signal debounce = 8/f_S 0011: Left AGC signal debounce = 16/f_S 0100: Left AGC signal debounce = 32/f_S 0101: Left AGC signal debounce = 64/f_S 0110: Left AGC signal debounce = 128/f_S 0111: Left AGC signal debounce = 256/f_S 1000: Left AGC signal debounce = 512/f_S 1001: Left AGC signal debounce = 1024/f_S 1010: Left AGC signal debounce = 2048/f_S 1011: Left AGC signal debounce = 2 × 2048/f_S 1100: Left AGC signal debounce = 3 × 2048/f_S 1101: Left AGC signal debounce = 4 × 2048/f_S 1110: Left AGC signal debounce = 5 × 2048/f_S 1111: Left AGC signal debounce = 6 × 2048/f_S

Table 71. Page 0 / Register 93: Left AGC Gain Applied

BIT(1)	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	0000 0000	Left AGC Gain Value Status: 1110 1000: Gain applied by left AGC = –12 dB 1110 1001: Gain applied by left AGC = –11.5 dB ... 1111 1111: Gain applied by left AGC = –0.5 dB 0000 0000: Gain applied by left AGC = 0 dB 0000 0001: Gain applied by left AGC = 0.5 dB ... 0100 1111: Gain applied by left AGC = 39.5 dB 0101 0000: Gain applied by left AGC = 40 dB 0101 0001–1111 1111: Reserved. Do not use.

(1) These are read-only bits.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 72. Page 0 / Register 94: Right AGC Control 1

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	0: Right AGC disabled 1: Right AGC enabled
D6–D4	R/W	000	000: Right AGC target level = –5.5 dB 000: Right AGC target level = –8 dB 001: Right AGC target level = –10 dB 010: Right AGC target level = –12 dB 011: Right AGC target level = –14 dB 100: Right AGC target level = –17 dB 101: Right AGC target level = –20 dB 111: Right AGC target level = –24 dB
D3–D0	R	0000	Reserved. Do not write any value other than reset value.

Table 73. Page 0 / Register 95: Right AGC Control 2

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D6	R/W	00	00: Right AGC hysteresis setting of 1 dB 01: Right AGC hysteresis setting of 2 dB 10: Right AGC hysteresis setting of 4 dB 11: Right AGC hysteresis disabled.
D5–D1	R/W	00 000	00 000: Right AGC noise/silence detection is disabled. 00 001: Right AGC noise threshold = –30 dB 00 010: Right AGC noise threshold = –32 dB 00 011: Right AGC noise threshold = –34 dB ... 11 101: Right AGC noise threshold = –86 dB 11 110: Right AGC noise threshold = –88 dB 11 111: Right AGC noise threshold = –90 dB
D0	R/W	0	0: Disable clip stepping for right AGC. 1: Enable clip stepping for right AGC.

Table 74. Page 0 / Register 96: Right AGC Maximum Gain

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R	0	Reserved. Do not write any value other than reset value.
D6–D0	R/W	111 1111	000 0000: Right AGC maximum gain = 0 dB 000 0001: Right AGC maximum gain = 0.5 dB 000 0010: Right AGC maximum gain = 1 dB ... 101 0000: Right AGC maximum gain = 40 dB 101 0001–111 1111: Not Used.

Table 75. Page 0 / Register 97: Right AGC Attack Time

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D3	R/W	0000 0	0000 0: Right AGC attack time = 1 × (32/f_S) 0000 1: Right AGC attack time = 3 × (32/f_S) 0001 0: Right AGC attack time = 5 × (32/f_S) 0001 1: Right AGC attack time = 7 × (32/f_S) 0010 0: Right AGC attack time = 9 × (32/f_S) ... 1111 0: Right AGC attack time = 61 × (32/f_S) 1111 1: Right AGC attack time = 63 × (32/f_S)
D2–D0	R/W	000	000: Multiplication factor for the programmed right AGC attack time = 1 001: Multiplication factor for the programmed right AGC attack time = 2 010: Multiplication factor for the programmed right AGC attack time = 4 ... 111: Multiplication factor for the programmed right AGC attack time = 128

Copyright © 2008–2015, Texas Instruments Incorporated

Table 76. Page 0 / Register 98: Right AGC Decay Time

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D3	R/W	0000 0	0000 0: Right AGC decay time = 1 × (512/f_S) 0000 1: Right AGC decay time = 3 × (512/f_S) 0001 0: Right AGC decay time = 5 × (512/f_S) 0001 1: Right AGC decay time = 7 × (512/f_S) 0010 0: Right AGC decay time = 9 × (512/f_S) ... 1111 0: Right AGC decay time = 61 × (512/f_S) 1111 1: Right AGC decay time = 63 × (512/f_S)
D2–D0	R/W	000	000: Multiply factor for the programmed right AGC decay time = 1 001: Multiply factor for the programmed right AGC decay time = 2 010: Multiply factor for the programmed right AGC decay time = 4 111: Multiply factor for the programmed right AGC decay time = 128

Table 77. Page 0 / Register 99: Right AGC Noise Debounce

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D5	R	000	Reserved. Do not write any value other than reset value.
D4–D0	R/W	0 0000	0 0000: Right AGC noise debounce = 0/f_S 0 0001: Right AGC noise debounce = 4/f_S 0 0010: Right AGC noise debounce = 8/f_S 0 0011: Right AGC noise debounce = 16/f_S 0 0100: Right AGC noise debounce = 32/f_S 0 0101: Right AGC noise debounce = 64/f_S 0 0110: Right AGC noise debounce = 128/f_S 0 0111: Right AGC noise debounce = 256/f_S 0 1000: Right AGC noise debounce = 512/f_S 0 1001: Right AGC noise debounce = 1024/f_S 0 1010: Right AGC noise debounce = 2048/f_S 0 1011: Right AGC noise debounce = 4096/f_S 0 1100: Right AGC noise debounce = 2 × 4096/f_S 0 1101: Right AGC noise debounce = 3 × 4096/f_S 0 1110: Right AGC noise debounce = 4 × 4096/f_S ... 1 1110: Right AGC noise debounce = 20 × 4096/f_S 1 1111: Right AGC noise debounce = 21 × 4096/f_S

Table 78. Page 0 / Register 100: Right AGC Signal Debounce

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R	0000	Reserved. Do not write any value other than reset value.
D3–D0	R/W	0000	0000: Right AGC signal debounce = 0/f_S 0001: Right AGC signal debounce = 4/f_S 0010: Right AGC signal debounce = 8/f_S 0011: Right AGC signal debounce = 16/f_S 0100: Right AGC signal debounce = 32/f_S 0101: Right AGC signal debounce = 64/f_S 0110: Right AGC signal debounce = 128/f_S 0111: Right AGC signal debounce = 256/f_S 1000: Right AGC signal debounce = 512/f_S 1001: Right AGC signal debounce = 1024/f_S 1010: Right AGC signal debounce = 2048/f_S 1011: Right AGC signal debounce = 2 × 2048/f_S 1100: Right AGC signal debounce = 3 × 2048/f_S 1101: Right AGC signal debounce = 4 × 2048/f_S 1110: Right AGC signal debounce = 5 × 2048/f_S 1111: Right AGC signal debounce = 6 × 2048/f_S

Copyright © 2008–2015, Texas Instruments Incorporated

Table 79. Page 0 / Register 101: Right AGC Gain Applied

BIT(1)	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	0000 0000	Right AGC Gain Value Status: 1110 1000: Gain applied by right AGC = –12 dB 1110 1001: Gain applied by right AGC = –11.5 dB ... 1111 1111: Gain applied by right AGC = –0.5 dB 0000 0000: Gain applied by right AGC = 0 dB 0000 0001: Gain applied by right AGC = 0.5 dB ... 0100 1111: Gain applied by right AGC = 39.5 dB 0101 0000: Gain applied by right AGC = 40 dB 0101 0001–1111 1111: Reserved. Do not use.

(1) These are read-only bits.

Table 80. Page 0 / Register 102 Through Page 0 / Register 127: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

10.6.3 CONTROL REGISTERS Page 1: ADC Routing, PGA, Power-Controls, Etc.

Table 81. Page 1 / Register 0: Page Control Register(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Page 0 selected 0000 0001: Page 1 selected ... 1111 1110: Page 254 selected (reserved) 1111 1111: Page 255 selected (reserved)

(1) Valid pages are 0, 1, 4, 5, 32–47. All other pages are reserved (do not access).

Table 82. Page 1 / Register 1 Through Page 1 / Register 25: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 83. Page 1 / Register 26: Dither Control

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R/W	0000	DC Offset Into Input of Left ADC; Signed Magnitude Number in ±15-mV Steps 1111: –105 mV ... 1011: –45 mV 1010: –30 mV 1001: –15 mV 0000: 0 mV 0001: 15 mV 0010: 30 mV 0011: 45 mV ... 0111: 105 mV
D3–D0	R/W	0000	DC Offset Into Input of Right ADC; Signed Magnitude Number in ±15-mV Steps 1111: –105 mV ... 1011: –45 mV 1010: –30 mV 1001: –15 mV 0000: 0 mV 0001: 15 mV 0010: 30 mV 0011: 45 mV ... 0111: 105 mV

Table 84. Page 1 / Register 27 Through Page 1 / Register 50: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Table 85. Page 1 / Register 51: MICBIAS Control

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D5	R	000	Reserved. Do not write any value other than reset value.
D4–D3	R/W	00	00: MICBIAS2 is powered down. 01: MICBIAS2 is powered to 2 V. 10: MICBIAS2 is powered to 2.5 V. 11: MICBIAS2 is connected to AVDD.
D2–D0	R	000	Reserved. Do not write any value other than reset value.

Table 86. Page 1 / Register 52: Left ADC Input Selection for Left PGA

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R/W	1111	Reserved. Do not write any value other than reset value.
D3–D2	R/W	11	IN2L Pin (Single-Ended)(1) 00: 0-dB setting is chosen. 01: –6-dB setting is chosen. 10: Is not connected to the left ADC PGA 11: Is not connected to the left ADC PGA
D1–D0	R/W	11	IN1l(P) Pin (Single-Ended)(1) 00: 0-dB setting is chosen. 01: –6-dB setting is chosen. 10: Is not connected to the left ADC PGA 11: Is not connected to the left ADC PGA

(1) To maintain the same PGA output level for both single-ended and differential pairs, the single-ended inputs have a 2× gain applied.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 87. Page 1 / Register 53: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 88. Page 1 / Register 54: Left ADC Input Selection for Left PGA

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	Reserved. Do not write any value other than reset value.
D6	R/W	0	Left ADC Common-Mode Select 0: Left ADC channel unselected inputs are not biased weakly to the ADC common-mode voltage. 1: Left ADC channel unselected inputs are biased weakly to the ADC common-mode voltage.
D5–D4	R/W	11	Differential Pair [Plus = IN1L(P) and Minus = IN1R(M)] 00: 0-dB setting is chosen. 01: –6-dB setting is chosen. 10–11: Not connected to the left ADC PGA
D3–D2	R/W	11	Reserved. Do not write any value other than reset value.
D1–D0	R/W	11	IN1R(M) Pin (Single-Ended)(1) 00: 0 dB setting is chosen. 01: –6 dB setting is chosen. 10–11: Not connected to the left ADC PGA.

(1) To maintain the same PGA output level for both single-ended and differential pairs, the single-ended inputs have a 2× gain applied.

Table 89. Page 1 / Register 55: Right ADC Input selection for Right PGA

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D2	R/W	11	Reserved. Do not write any value other than reset value.
D1–D0	R/W	11	IN1R(M) Pin (Single-Ended)(1) 00: 0-dB setting is chosen. 01: –6-dB setting is chosen. 10–11: Not connected to the right ADC PGA.

(1) To maintain the same PGA output level for both single-ended and differential pairs, the single-ended inputs have a 2× gain applied.

Table 90. Page 1 / Register 56: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 91. Page 1 / Register 57: Right ADC Input Selection for Right PGA

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	0	Reserved. Do not write any value other than reset value.
D6	R/W	0	Right ADC Common-Mode Select 0: Right ADC channel unselected inputs are not biased weakly to the ADC common-mode voltage. 1: Right ADC channel unselected inputs are biased weakly to the ADC common-mode voltage.
D5–D4	R/W	11	Differential Pair [Plus = IN1L(P) and Minus = IN1R(M)] 00: 0-dB setting is chosen. 01: –6 dB setting is chosen. 10–11: Not connected to the right ADC PGA
D3–D2	R/W	11	Reserved. Do not write any value other than reset value.
D1–D0	R/W	11	IN1L(P) Pin (Single-Ended)(1) 00: 0-dB setting is chosen. 01: –6-dB setting is chosen. 10–11: Not connected to the right ADC PGA

(1) To maintain the same PGA output level for both single-ended and differential pairs, the single-ended inputs have a 2× gain applied.

Copyright © 2008–2015, Texas Instruments Incorporated

Table 92. Page 1 / Register 58: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to this register.

Table 93. Page 1 / Register 59: Left Analog PGA Settings

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	1	0: Left PGA is not muted. 1: Left PGA is muted.
D6–D0	R/W	000 0000	000 0000: Left PGA gain = 0 dB 000 0001: Left PGA gain = 0.5 dB 000 0010: Left PGA gain = 1 dB ... 101 0000: Left PGA gain = 40 dB 101 0001–111 1111: Reserved. Do not use.

Table 94. Page 1 / Register 60: Right Analog PGA Settings

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7	R/W	1	0: Right PGA is not muted 1: Right PGA is muted
D6–D0	R/W	000 0000	000 0000: Right PGA gain = 0 dB 000 0001: Right PGA gain = 0.5 dB 000 010: Right PGA gain = 1 dB ... 101 0000: Right PGA gain = 40 dB 101 0001–111 1111: Reserved. Do not use.

Table 95. Page 1 / Register 61: ADC Low-Current Modes

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D1	R	0000 000	Reserved. Write only zeros to these bits.
D0	R/W	0	0: 1× ADC modulator current used 1: 0.5× ADC modulator current used

Table 96. Page 1 / Register 62: ADC Analog PGA Flags

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D2	R	0000 00	Reserved. Do not write any value other than reset value.
D1	R	0	0: Left ADC PGA, applied gain ≠ programmed gain 1: Left ADC PGA, applied gain = programmed gain
D0	R	0	0: Right ADC PGA, applied gain ≠ programmed gain 1: Right ADC PGA, applied gain = programmed gain

Table 97. Page 1 / Register 63 Through Page 1 / Register 127: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R	XXXX XXXX	Reserved. Do not write to these registers.

Copyright © 2008–2015, Texas Instruments Incorporated

10.6.4 Control Registers, Page 4: ADC Digital Filter Coefficients

Default values shown for this page only become valid 100 µs following a hardware or software reset.

Table 98. Page 4 / Register 0: Page Control Register(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Page 0 selected 0000 0001: Page 1 selected ... 1111 1110: Page 254 selected (reserved) 1111 1111: Page 255 selected (reserved)

(1) Valid pages are 0, 1, 4, 5, 32–47. All other pages are reserved (do not access).

The remaining page-4 registers are either reserved registers or are used for setting coefficients for the various filters in the processing blocks. Reserved registers must not be written to.

The filter coefficient registers are arranged in pairs, with two adjacent 8-bit registers containing the 16-bit coefficient for a single filter. The 16-bit integer contained in the MSB and LSB registers for a coefficient are interpreted as a 2s-complement integer, with possible values ranging from –32,768 to 32,767. When programming any coefficient value for a filter, the MSB register must always be written first, immediately followed by the LSB register. Even if only the MSB or LSB portion of the coefficient changes, both registers must be written in this sequence. Table 99 is a list of the page-4 registers, excepting the previously described register 0.

Table 99. Page-4 Registers

REGISTER NUMBER	RESET VALUE	REGISTER NAME
1	XXXX XXXX	Reserved. Do not write to this register.
2	0000 0001	Coefficient N0(15:8) for AGC LPF (first-order IIR) used as averager to detect level or coefficient C1(15:8) of ADC miniDSP
3	0001 0111	Coefficient N0(7:0) for AGC LPF (first-order IIR) used as averager to detect level or coefficient C1(7:0) of ADC miniDSP
4	0000 0001	Coefficient N1(15:8) for AGC LPF (first-order IIR) used as averager to detect level or coefficient C2(15:8) of ADC miniDSP
5	0001 0111	Coefficient N1(7:0) for AGC LPF (first-order IIR) used as averager to detect level or coefficient C2(7:0) of ADC miniDSP
6	0111 1101	Coefficient D1(15:8) for AGC LPF (first-order IIR) used as averager to detect level or coefficient C3(15:8) of ADC miniDSP
7	1101 0011	Coefficient D1(7:0) for AGC LPF (first-order IIR) used as averager to detect level or coefficient C3(7:0) of ADC miniDSP
8	0111 1111	Coefficient N0(15:8) for left ADC programmable first-order IIR or coefficient C4(15:8) of ADC miniDSP
9	1111 1111	Coefficient N0(7:0) for left ADC programmable first-order IIR or coefficient C4(7:0) of ADC miniDSP
10	0000 0000	Coefficient N1(15:8) for left ADC programmable first-order IIR or coefficient C5(15:8) of ADC miniDSP
11	0000 0000	Coefficient N1(7:0) for left ADC programmable first-order IIR or coefficient C5(7:0) of ADC miniDSP
12	0000 0000	Coefficient D1(15:8) for left ADC programmable first-order IIR or coefficient C6(15:8) of ADC miniDSP
13	0000 0000	Coefficient D1(7:0) for left ADC programmable first-order IIR or coefficient C6(7:0) of ADC miniDSP
14	0111 1111	Coefficient N0(15:8) for left ADC biquad A or coefficient FIR0(15:8) for ADC FIR filter or coefficient C7(15:8) of ADC miniDSP
15	1111 1111	Coefficient N0(7:0) for left ADC biquad A or coefficient FIR0(7:0) for ADC FIR filter or coefficient C7(7:0) of ADC miniDSP
16	0000 0000	Coefficient N1(15:8) for left ADC biquad A or coefficient FIR1(15:8) for ADC FIR filter or coefficient C8(15:8) of ADC miniDSP
17	0000 0000	Coefficient N1(7:0) for left ADC biquad A or coefficient FIR1(7:0) for ADC FIR filter or coefficient C8(7:0) of ADC miniDSP
18	0000 0000	Coefficient N2(15:8) for left ADC biquad A or coefficient FIR2(15:8) for ADC FIR filter or coefficient C9(15:8) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

Table 99. Page-4 Registers (continued)

REGISTER NUMBER	RESET VALUE	REGISTER NAME
19	0000 0000	Coefficient N2(7:0) for left ADC biquad A or coefficient FIR2(7:0) for ADC FIR filter or coefficient C9(7:0) of ADC miniDSP
20	0000 0000	Coefficient D1(15:8) for left ADC biquad A or coefficient FIR3(15:8) for ADC FIR filter or coefficient C10(15:8) of ADC miniDSP
21	0000 0000	Coefficient D1(7:0) for left ADC biquad A or coefficient FIR3(7:0) for ADC FIR filter or coefficient C10(7:0) of ADC miniDSP
22	0000 0000	Coefficient D2(15:8) for left ADC biquad A or coefficient FIR4(15:8) for ADC FIR filter or coefficient C11(15:8) of ADC miniDSP
23	0000 0000	Coefficient D2(7:0) for left ADC biquad A or coefficient FIR4(7:0) for ADC FIR filter or coefficient C11(7:0) of ADC miniDSP
24	0111 1111	Coefficient N0(15:8) for left ADC biquad B or coefficient FIR5(15:8) for ADC FIR filter or coefficient C12(15:8) of ADC miniDSP
25	1111 1111	Coefficient N0(7:0) for left ADC biquad B or coefficient FIR5(7:0) for ADC FIR filter or coefficient C12(7:0) of ADC miniDSP
26	0000 0000	Coefficient N1(15:8) for left ADC biquad B or coefficient FIR6(15:8) for ADC FIR filter or coefficient C13(15:8) of ADC miniDSP
27	0000 0000	Coefficient N1(7:0) for left ADC biquad B or coefficient FIR6(7:0) for ADC FIR filter or coefficient C13(7:0) of ADC miniDSP
28	0000 0000	Coefficient N2(15:8) for left ADC biquad B or coefficient FIR7(15:8) for ADC FIR filter or coefficient C14(15:8) of ADC miniDSP
29	0000 0000	Coefficient N2(7:0) for left ADC biquad B or coefficient FIR7(7:0) for ADC FIR filter or coefficient C14(7:0) of ADC miniDSP
30	0000 0000	Coefficient D1(15:8) for left ADC biquad B or coefficient FIR8(15:8) for ADC FIR filter or coefficient C15(15:8) of ADC miniDSP
31	0000 0000	Coefficient D1(7:0) for left ADC biquad B or coefficient FIR8(7:0) for ADC FIR filter or coefficient C15(7:0) of ADC miniDSP
32	0000 0000	Coefficient D2(15:8) for left ADC biquad B or coefficient FIR9(15:8) for ADC FIR filter or coefficient C16(15:8) of ADC miniDSP
33	0000 0000	Coefficient D2(7:0) for left ADC biquad B or coefficient FIR9(7:0) for ADC FIR filter or coefficient C16(7:0) of ADC miniDSP
34	0111 1111	Coefficient N0(15:8) for left ADC biquad C or coefficient FIR10(15:8) for ADC FIR filter or coefficient C17(15:8) of ADC miniDSP
35	1111 1111	Coefficient N0(7:0) for left ADC biquad C or coefficient FIR10(7:0) for ADC FIR filter or coefficient C17(7:0) of ADC miniDSP
36	0000 0000	Coefficient N1(15:8) for left ADC biquad C or coefficient FIR11(15:8) for ADC FIR filter or coefficient C18(15:8) of ADC miniDSP
37	0000 0000	Coefficient N1(7:0) for left ADC biquad C or coefficient FIR11(7:0) for ADC FIR filter or coefficient C18(7:0) of ADC miniDSP
38	0000 0000	Coefficient N2(15:8) for left ADC biquad C or coefficient FIR12(15:8) for ADC FIR filter or coefficient C19(15:8) of ADC miniDSP
39	0000 0000	Coefficient N2(7:0) for left ADC biquad C or coefficient FIR12(7:0) for ADC FIR filter or coefficient C19(7:0) of ADC miniDSP
40	0000 0000	Coefficient D1(15:8) for left ADC biquad C or coefficient FIR13(15:8) for ADC FIR filter or coefficient C20(15:8) of ADC miniDSP
41	0000 0000	Coefficient D1(7:0) for left ADC biquad C or coefficient FIR13(7:0) for ADC FIR filter or coefficient C20(7:0) of ADC miniDSP
42	0000 0000	Coefficient D2(15:8) for left ADC biquad C or coefficient FIR14(15:8) for ADC FIR filter or coefficient C21(15:8) of ADC miniDSP
43	0000 0000	Coefficient D2(7:0) for left ADC biquad C or coefficient FIR14(7:0) for ADC FIR filter or coefficient C21(7:0) of ADC miniDSP
44	0111 1111	Coefficient N0(15:8) for left ADC biquad D or coefficient FIR15(15:8) for ADC FIR filter or coefficient C22(15:8) of ADC miniDSP
45	1111 1111	Coefficient N0(7:0) for left ADC biquad D or coefficient FIR15(7:0) for ADC FIR filter or coefficient C22(7:0) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

Table 99. Page-4 Registers (continued)

REGISTER NUMBER	RESET VALUE	REGISTER NAME
46	0000 0000	Coefficient N1(15:8) for left ADC biquad D or coefficient FIR16(15:8) for ADC FIR filter or coefficient C23(15:8) of ADC miniDSP
47	0000 0000	Coefficient N1(7:0) for left ADC biquad D or coefficient FIR16(7:0) for ADC FIR filter or coefficient C23(7:0) of ADC miniDSP
48	0000 0000	Coefficient N2(15:8) for left ADC biquad D or coefficient FIR17(15:8) for ADC FIR filter or coefficient C24(15:8) of ADC miniDSP
49	0000 0000	Coefficient N2(7:0) for left ADC biquad D or coefficient FIR17(7:0) for ADC FIR filter or coefficient C24(7:0) of ADC miniDSP
50	0000 0000	Coefficient D1(15:8) for left ADC biquad D or coefficient FIR18(15:8) for ADC FIR filter or coefficient C25(15:8) of ADC miniDSP
51	0000 0000	Coefficient D1(7:0) for left ADC biquad D or coefficient FIR18(7:0) for ADC FIR filter or coefficient C25(7:0) of ADC miniDSP
52	0000 0000	Coefficient D2(15:8) for left ADC biquad D or coefficient FIR19(15:8) for ADC FIR filter or coefficient C26(15:8) of ADC miniDSP
53	0000 0000	Coefficient D2(7:0) for left ADC biquad D or coefficient FIR19(7:0) for ADC FIR filter or coefficient C26(7:0) of ADC miniDSP
54	0111 1111	Coefficient N0(15:8) for left ADC biquad E or coefficient FIR20(15:8) for ADC FIR filter or coefficient C27(15:8) of ADC miniDSP
55	1111 1111	Coefficient N0(7:0) for left ADC biquad E or coefficient FIR20(7:0) for ADC FIR filter or coefficient C27(7:0) of ADC miniDSP
56	0000 0000	Coefficient N1(15:8) for left ADC biquad E or coefficient FIR21(15:8) for ADC FIR filter or coefficient C28(15:8) of ADC miniDSP
57	0000 0000	Coefficient N1(7:0) for left ADC biquad E or coefficient FIR21(7:0) for ADC FIR filter or coefficient C28(7:0) of ADC miniDSP
58	0000 0000	Coefficient N2(15:8) for left ADC biquad E or coefficient FIR22(15:8) for ADC FIR filter or coefficient C29(15:8) of ADC miniDSP
59	0000 0000	Coefficient N2(7:0) for left ADC biquad E or coefficient FIR22(7:0) for ADC FIR filter or coefficient C29(7:0) of ADC miniDSP
60	0000 0000	Coefficient D1(15:8) for left ADC biquad E or coefficient FIR23(15:8) for ADC FIR filter or coefficient C30(15:8) of ADC miniDSP
61	0000 0000	Coefficient D1(7:0) for left ADC biquad E or coefficient FIR23(7:0) for ADC FIR filter or coefficient C30(7:0) of ADC miniDSP
62	0000 0000	Coefficient D2(15:8) for left ADC biquad E or coefficient FIR24(15:8) for ADC FIR filter or coefficient C31(15:8) of ADC miniDSP
63	0000 0000	Coefficient D2(7:0) for left ADC biquad E or coefficient FIR24(7:0) for ADC FIR filter or coefficient C31(7:0) of ADC miniDSP
64	0000 0000	Coefficient C32(15:8) of ADC miniDSP
65	0000 0000	Coefficient C32(7:0) of ADC miniDSP
66	0000 0000	Coefficient C33(15:8) of ADC miniDSP
67	0000 0000	Coefficient C33(7:0) of ADC miniDSP
68	0000 0000	Coefficient C34(15:8) of ADC miniDSP
69	0000 0000	Coefficient C34(7:0) of ADC miniDSP
70	0000 0000	Coefficient C35(15:8) of ADC miniDSP
71	0000 0000	Coefficient C35(7:0) of ADC miniDSP
72	0000 0000	Coefficient N0(15:8) for right ADC programmable first-order IIR or coefficient C36(15:8) of ADC miniDSP
73	0000 0000	Coefficient N0(7:0) for right ADC programmable first-order IIR or coefficient C36(7:0) of ADC miniDSP
74	0000 0000	Coefficient N1(15:8) for right ADC programmable first-order IIR or coefficient C37(15:8) of ADC miniDSP
75	0000 0000	Coefficient N1(7:0) for right ADC programmable first-order IIR or coefficient C37(7:0) of ADC miniDSP
76	0000 0000	Coefficient D1(15:8) for right ADC programmable first-order IIR or coefficient C38(15:8) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

Table 99. Page-4 Registers (continued)

REGISTER NUMBER	RESET VALUE	REGISTER NAME
77	0000 0000	Coefficient D1(7:0) for right ADC programmable first-order IIR or coefficient C38(7:0) of ADC miniDSP
78	0000 0000	Coefficient N0(15:8) for right ADC biquad A or coefficient FIR0(15:8) for ADC FIR filter or coefficient C39(15:8) of ADC miniDSP
79	0000 0000	Coefficient N0(7:0) for right ADC biquad A or coefficient FIR0(7:0) for ADC FIR filter or coefficient C39(7:0) of ADC miniDSP
80	0000 0000	Coefficient N1(15:8) for right ADC biquad A or coefficient FIR1(15:8) for ADC FIR filter or coefficient C40(15:8) of ADC miniDSP
81	0000 0000	Coefficient N1(7:0) for right ADC biquad A or coefficient FIR1(7:0) for ADC FIR filter or coefficient C40(7:0) of ADC miniDSP
82	0000 0000	Coefficient N2(15:8) for right ADC biquad A or coefficient FIR2(15:8) for ADC FIR filter or coefficient C41(15:8) of ADC miniDSP
83	0000 0000	Coefficient N2(7:0) for right ADC biquad A or coefficient FIR2(7:0) for ADC FIR filter or coefficient C41(7:0) of ADC miniDSP
84	0000 0000	Coefficient D1(15:8) for right ADC biquad A or coefficient FIR3(15:8) for ADC FIR filter or coefficient C42(15:8) of ADC miniDSP
85	0000 0000	Coefficient D1(7:0) for right ADC biquad A or coefficient FIR3(7:0) for ADC FIR filter or coefficient C42(7:0) of ADC miniDSP
86	0000 0000	Coefficient D2(15:8) for right ADC biquad A or coefficient FIR4(15:8) for ADC FIR filter or coefficient C43(15:8) of ADC miniDSP
87	0000 0000	Coefficient D2(7:0) for right ADC biquad A or coefficient FIR4(7:0) for ADC FIR filter or coefficient C43(7:0) of ADC miniDSP
88	0000 0000	Coefficient N0(15:8) for right ADC biquad B or coefficient FIR5(15:8) for ADC FIR filter or coefficient C44(15:8) of ADC miniDSP
89	0000 0000	Coefficient N0(7:0) for right ADC biquad B or coefficient FIR5(7:0) for ADC FIR filter or coefficient C44(7:0) of ADC miniDSP
90	0000 0000	Coefficient N1(15:8) for right ADC biquad B or coefficient FIR6(15:8) for ADC FIR filter or coefficient C45(15:8) of ADC miniDSP
91	0000 0000	Coefficient N1(7:0) for right ADC biquad B or coefficient FIR6(7:0) for ADC FIR filter or coefficient C45(7:0) of ADC miniDSP
92	0000 0000	Coefficient N2(15:8) for right ADC biquad B or coefficient FIR7(15:8) for ADC FIR filter or coefficient C46(15:8) of ADC miniDSP
93	0000 0000	Coefficient N2(7:0) for right ADC biquad B or coefficient FIR7(7:0) for ADC FIR filter or coefficient C46(7:0) of ADC miniDSP
94	0000 0000	Coefficient D1(15:8) for right ADC biquad B or coefficient FIR8(15:8) for ADC FIR filter or coefficient C47(15:8) of ADC miniDSP
95	0000 0000	Coefficient D1(7:0) for right ADC biquad B or coefficient FIR8(7:0) for ADC FIR filter or coefficient C47(7:0) of ADC miniDSP
96	0000 0000	Coefficient D2(15:8) for right ADC biquad B or coefficient FIR9(15:8) for ADC FIR filter or coefficient C48(15:8) of ADC miniDSP
97	0000 0000	Coefficient D2(7:0) for right ADC biquad B or coefficient FIR9(7:0) for ADC FIR filter or coefficient C48(7:0) of ADC miniDSP
98	0000 0000	Coefficient N0(15:8) for right ADC biquad C or coefficient FIR10(15:8) for ADC FIR filter or coefficient C49(15:8) of ADC miniDSP
99	0000 0000	Coefficient N0(7:0) for right ADC biquad C or coefficient FIR10(7:0) for ADC FIR filter or coefficient C49(7:0) of ADC miniDSP
100	0000 0000	Coefficient N1(15:8) for right ADC biquad C or coefficient FIR11(15:8) for ADC FIR filter or coefficient C50(15:8) of ADC miniDSP
101	0000 0000	Coefficient N1(7:0) for right ADC biquad C or coefficient FIR11(7:0) for ADC FIR filter or coefficient C50(7:0) of ADC miniDSP
102	0000 0000	Coefficient N2(15:8) for right ADC biquad C or coefficient FIR12(15:8) for ADC FIR filter or coefficient C51(15:8) of ADC miniDSP
103	0000 0000	Coefficient N2(7:0) for right ADC biquad C or coefficient FIR12(7:0) for ADC FIR filter or coefficient C51(7:0) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

Table 99. Page-4 Registers (continued)

REGISTER NUMBER	RESET VALUE	REGISTER NAME
104	0000 0000	Coefficient D1(15:8) for right ADC biquad C or coefficient FIR13(15:8) for ADC FIR filter or coefficient C52(15:8) of ADC miniDSP
105	0000 0000	Coefficient D1(7:0) for right ADC biquad C or coefficient FIR13(7:0) for ADC FIR filter or coefficient C52(7:0) of ADC miniDSP
106	0000 0000	Coefficient D2(15:8) for right ADC biquad C or coefficient FIR14(15:8) for ADC FIR filter or coefficient C53(15:8) of ADC miniDSP
107	0000 0000	Coefficient D2(7:0) for right ADC biquad C or coefficient FIR14(7:0) for ADC FIR filter or coefficient C53(7:0) of ADC miniDSP
108	0000 0000	Coefficient N0(15:8) for right ADC biquad D or coefficient FIR15(15:8) for ADC FIR filter or coefficient C54(15:8) of ADC miniDSP
109	0000 0000	Coefficient N0(7:0) for right ADC biquad D or coefficient FIR15(7:0) for ADC FIR filter or coefficient C54(7:0) of ADC miniDSP
110	0000 0000	Coefficient N1(15:8) for right ADC biquad D or coefficient FIR16(15:8) for ADC FIR filter or coefficient C55(15:8) of ADC miniDSP
111	0000 0000	Coefficient N1(7:0) for right ADC biquad D or coefficient FIR16(7:0) for ADC FIR filter or coefficient C55(7:0) of ADC miniDSP
112	0000 0000	Coefficient N2(15:8) for right ADC biquad D or coefficient FIR17(15:8) for ADC FIR filter or coefficient C56(15:8) of ADC miniDSP
113	0000 0000	Coefficient N2(7:0) for right ADC biquad D or coefficient FIR17(7:0) for ADC FIR filter or coefficient C56(7:0) of ADC miniDSP
114	0000 0000	Coefficient D1(15:8) for right ADC biquad D or coefficient FIR18(15:8) for ADC FIR filter or coefficient C57(15:8) of ADC miniDSP
115	0000 0000	Coefficient D1(7:0) for right ADC biquad D or coefficient FIR18(7:0) for ADC FIR filter or coefficient C57(7:0) of ADC miniDSP
116	0000 0000	Coefficient D2(15:8) for right ADC biquad D or coefficient FIR19(15:8) for ADC FIR filter or coefficient C58(15:8) of ADC miniDSP
117	0000 0000	Coefficient D2(7:0) for right ADC biquad D or coefficient FIR19(7:0) for ADC FIR filter or coefficient C58(7:0) of ADC miniDSP
118	0000 0000	Coefficient N0(15:8) for right ADC biquad E or coefficient FIR20(15:8) for ADC FIR filter or coefficient C59(15:8) of ADC miniDSP
119	0000 0000	Coefficient N0(7:0) for right ADC biquad E or coefficient FIR20(7:0) for ADC FIR filter or coefficient C59(7:0) of ADC miniDSP
120	0000 0000	Coefficient N1(15:8) for right ADC biquad E or coefficient FIR21(15:8) for ADC FIR filter or coefficient C60(15:8) of ADC miniDSP
121	0000 0000	Coefficient N1(7:0) for right ADC biquad E or coefficient FIR21(7:0) for ADC FIR filter or coefficient C60(7:0) of ADC miniDSP
122	0000 0000	Coefficient N2(15:8) for right ADC biquad E or coefficient FIR22(15:8) for ADC FIR filter or coefficient C61(15:8) of ADC miniDSP
123	0000 0000	Coefficient N2(7:0) for right ADC biquad E or coefficient FIR22(7:0) for ADC FIR filter or coefficient C61(7:0) of ADC miniDSP
124	0000 0000	Coefficient D1(15:8) for right ADC biquad E or coefficient FIR23(15:8) for ADC FIR filter or coefficient C62(15:8) of ADC miniDSP
125	0000 0000	Coefficient D1(7:0) for right ADC biquad E or coefficient FIR23(7:0) for ADC FIR filter or coefficient C62(7:0) of ADC miniDSP
126	0000 0000	Coefficient D2(15:8) for right ADC biquad E or coefficient FIR24(15:8) for ADC FIR filter or coefficient C63(15:8) of ADC miniDSP
127	0000 0000	Coefficient D2(7:0) for right ADC biquad E or coefficient FIR24(7:0) for ADC FIR filter or coefficient C63(7:0) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

10.6.5 Control Registers, Page 5: ADC Programmable Coefficients RAM (65:127)

Page 5 / register 0 is the page control register as desribed following.

Table 100. Page 5 / Register 0: Page Control Register(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Page 0 selected 0000 0001: Page 1 selected ... 1111 1110: Page 254 selected (reserved) 1111 1111: Page 255 selected (reserved)

(1) Valid pages are 0, 1, 4, 5, and 32–47. All other pages are reserved (do not access).

Table 101is a list of the page-5 registers, excepting the previously described register 0.

Table 101. Page-5 Registers

REGISTER NUMBER	RESET VALUE	REGISTER NAME
1	XXXX XXXX	Reserved. Do not write to this register.
2	0000 0000	Coefficient C65(15:8) of ADC miniDSP
3	0000 0000	Coefficient C65(7:0) of ADC miniDSP
4	0000 0000	Coefficient C66(15:8) of ADC miniDSP
5	0000 0000	Coefficient C66(7:0) of ADC miniDSP
6	0000 0000	Coefficient C67(15:8) of ADC miniDSP
7	0000 0000	Coefficient C67(7:0) of ADC miniDSP
8	0000 0000	Coefficient C68(15:8) of ADC miniDSP
9	0000 0000	Coefficient C68(7:0) of ADC miniDSP
10	0000 0000	Coefficient C69(15:8) of ADC miniDSP
11	0000 0000	Coefficient C69(7:0) of ADC miniDSP
12	0000 0000	Coefficient C70(15:8) of ADC miniDSP
13	0000 0000	Coefficient C70(7:0) of ADC miniDSP
14	0000 0000	Coefficient C71(15:8) of ADC miniDSP
15	0000 0000	Coefficient C71(7:0) of ADC miniDSP
16	0000 0000	Coefficient C72(15:8) of ADC miniDSP
17	0000 0000	Coefficient C72(7:0) of ADC miniDSP
18	0000 0000	Coefficient C73(15:8) of ADC miniDSP
19	0000 0000	Coefficient C73(7:0) of ADC miniDSP
20	0000 0000	Coefficient C74(15:8) of ADC miniDSP
21	0000 0000	Coefficient C74(7:0) of ADC miniDSP
22	0000 0000	Coefficient C75(15:8) of ADC miniDSP
23	0000 0000	Coefficient C75(7:0) of ADC miniDSP
24	0000 0000	Coefficient C76(15:8) of ADC miniDSP
25	0000 0000	Coefficient C76(7:0) of ADC miniDSP
26	0000 0000	Coefficient C77(15:8) of ADC miniDSP
27	0000 0000	Coefficient C77(7:0) of ADC miniDSP
28	0000 0000	Coefficient C78(15:8) of ADC miniDSP
29	0000 0000	Coefficient C78(7:0) of ADC miniDSP
30	0000 0000	Coefficient C79(15:8) of ADC miniDSP
31	0000 0000	Coefficient C79(7:0) of ADC miniDSP
32	0000 0000	Coefficient C80(15:8) of ADC miniDSP
33	0000 0000	Coefficient C80(7:0) of ADC miniDSP
34	0000 0000	Coefficient C81(15:8) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

Table 101. Page-5 Registers (continued)

REGISTER NUMBER	RESET VALUE	REGISTER NAME
35	0000 0000	Coefficient C81(7:0) of ADC miniDSP
36	0000 0000	Coefficient C82(15:8) of ADC miniDSP
37	0000 0000	Coefficient C82(7:0) of ADC miniDSP
38	0000 0000	Coefficient C83(15:8) of ADC miniDSP
39	0000 0000	Coefficient C83(7:0) of ADC miniDSP
40	0000 0000	Coefficient C84(15:8) of ADC miniDSP
41	0000 0000	Coefficient C84(7:0) of ADC miniDSP
42	0000 0000	Coefficient C85(15:8) of ADC miniDSP
43	0000 0000	Coefficient C85(7:0) of ADC miniDSP
44	0000 0000	Coefficient C86(15:8) of ADC miniDSP
45	0000 0000	Coefficient C86(7:0) of ADC miniDSP
46	0000 0000	Coefficient C87(15:8) of ADC miniDSP
47	0000 0000	Coefficient C87(7:0) of ADC miniDSP
48	0000 0000	Coefficient C88(15:8) of ADC miniDSP
49	0000 0000	Coefficient C88(7:0) of ADC miniDSP
50	0000 0000	Coefficient C89(15:8) of ADC miniDSP
51	0000 0000	Coefficient C89(7:0) of ADC miniDSP
52	0000 0000	Coefficient C90(15:8) of ADC miniDSP
53	0000 0000	Coefficient C90(7:0) of ADC miniDSP
54	0000 0000	Coefficient C91(15:8) of ADC miniDSP
55	0000 0000	Coefficient C91(7:0) of ADC miniDSP
56	0000 0000	Coefficient C92(15:8) of ADC miniDSP
57	0000 0000	Coefficient C92(7:0) of ADC miniDSP
58	0000 0000	Coefficient C93(15:8) of ADC miniDSP
59	0000 0000	Coefficient C93(7:0) of ADC miniDSP
60	0000 0000	Coefficient C94(15:8) of ADC miniDSP
61	0000 0000	Coefficient C94(7:0) of ADC miniDSP
62	0000 0000	Coefficient C95(15:8) of ADC miniDSP
63	0000 0000	Coefficient C95(7:0) of ADC miniDSP
64	0000 0000	Coefficient C96(15:8) of ADC miniDSP
65	0000 0000	Coefficient C96(7:0) of ADC miniDSP
66	0000 0000	Coefficient C97(15:8) of ADC miniDSP
67	0000 0000	Coefficient C97(7:0) of ADC miniDSP
68	0000 0000	Coefficient C98(15:8) of ADC miniDSP
69	0000 0000	Coefficient C98(7:0) of ADC miniDSP
70	0000 0000	Coefficient C99(15:8) of ADC miniDSP
71	0000 0000	Coefficient C99(7:0) of ADC miniDSP
72	0000 0000	Coefficient C100(15:8) of ADC miniDSP
73	0000 0000	Coefficient C100(7:0) of ADC miniDSP
74	0000 0000	Coefficient C101(15:8) of ADC miniDSP
75	0000 0000	Coefficient C101(7:0) of ADC miniDSP
76	0000 0000	Coefficient C102(15:8) of ADC miniDSP
77	0000 0000	Coefficient C102(7:0) of ADC miniDSP
78	0000 0000	Coefficient C103(15:8) of ADC miniDSP
79	0000 0000	Coefficient C103(7:0) of ADC miniDSP
80	0000 0000	Coefficient C104(15:8) of ADC miniDSP
81	0000 0000	Coefficient C104(7:0) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

Table 101. Page-5 Registers (continued)

REGISTER NUMBER	RESET VALUE	REGISTER NAME
82	0000 0000	Coefficient C105(15:8) of ADC miniDSP
83	0000 0000	Coefficient C105(7:0) of ADC miniDSP
84	0000 0000	Coefficient C106(15:8) of ADC miniDSP
85	0000 0000	Coefficient C106(7:0) of ADC miniDSP
86	0000 0000	Coefficient C107(15:8) of ADC miniDSP
87	0000 0000	Coefficient C107(7:0) of ADC miniDSP
88	0000 0000	Coefficient C108(15:8) of ADC miniDSP
89	0000 0000	Coefficient C108(7:0) of ADC miniDSP
90	0000 0000	Coefficient C109(15:8) of ADC miniDSP
91	0000 0000	Coefficient C109(7:0) of ADC miniDSP
92	0000 0000	Coefficient C110(15:8) of ADC miniDSP
93	0000 0000	Coefficient C110(7:0) of ADC miniDSP
94	0000 0000	Coefficient C111(15:8) of ADC miniDSP
95	0000 0000	Coefficient C111(7:0) of ADC miniDSP
96	0000 0000	Coefficient C112(15:8) of ADC miniDSP
97	0000 0000	Coefficient C112(7:0) of ADC miniDSP
98	0000 0000	Coefficient C113(15:8) of ADC miniDSP
99	0000 0000	Coefficient C113(7:0) of ADC miniDSP
100	0000 0000	Coefficient C114(15:8) of ADC miniDSP
101	0000 0000	Coefficient C114(7:0) of ADC miniDSP
102	0000 0000	Coefficient C115(15:8) of ADC miniDSP
103	0000 0000	Coefficient C115(7:0) of ADC miniDSP
104	0000 0000	Coefficient C117(15:8) of ADC miniDSP
105	0000 0000	Coefficient C117(7:0) of ADC miniDSP
106	0000 0000	Coefficient C117(15:8) of ADC miniDSP
107	0000 0000	Coefficient C117(7:0) of ADC miniDSP
108	0000 0000	Coefficient C118(15:8) of ADC miniDSP
109	0000 0000	Coefficient C118(7:0) of ADC miniDSP
110	0000 0000	Coefficient C119(15:8) of ADC miniDSP
111	0000 0000	Coefficient C119(7:0) of ADC miniDSP
112	0000 0000	Coefficient C120(15:8) of ADC miniDSP
113	0000 0000	Coefficient C120(7:0) of ADC miniDSP
114	0000 0000	Coefficient C121(15:8) of ADC miniDSP
115	0000 0000	Coefficient C121(7:0) of ADC miniDSP
116	0000 0000	Coefficient C122(15:8) of ADC miniDSP
117	0000 0000	Coefficient C122(7:0) of ADC miniDSP
118	0000 0000	Coefficient C123(15:8) of ADC miniDSP
119	0000 0000	Coefficient C123(7:0) of ADC miniDSP
120	0000 0000	Coefficient C124(15:8) of ADC miniDSP
121	0000 0000	Coefficient C124(7:0) of ADC miniDSP
122	0000 0000	Coefficient C125(15:8) of ADC miniDSP
123	0000 0000	Coefficient C125(7:0) of ADC miniDSP
124	0000 0000	Coefficient C126(15:8) of ADC miniDSP
125	0000 0000	Coefficient C126(7:0) of ADC miniDSP
126	0000 0000	Coefficient C127(15:8) of ADC miniDSP
127	0000 0000	Coefficient C127(7:0) of ADC miniDSP

Copyright © 2008–2015, Texas Instruments Incorporated

10.6.6 Control Registers, Page 32: ADC DSP Engine Instruction RAM (0:31)

Control registers from page 32 through page 47 contain instruction RAM for the ADC miniDSP. There are 32 instructions / page and 16 pages, so the TLV320ADC3001 miniDSP supports 512 instructions.

Table 102. Page 32 / Register 0: Page Control Register(1)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	0000 0000	0000 0000: Page 0 selected 0000 0001: Page 1 selected ... 1111 1110: Page 254 selected (reserved) 1111 1111: Page 255 selected (reserved)

(1) Valid pages are 0, 1, 4, 5, and 32–47. All other pages are reserved (do not access).

Table 103. Page 32 / Register 1: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	XXXX XXXX	Reserved. Write only the default value to this register

Table 104. Page 32 / Register 2: Inst_0(19:16)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D4	R/W	XXXX	Reserved
D3–D0	R/W	XXXX	Instruction Inst_0(19:16) of ADC miniDSP

Table 105. Page 32 / Register 3: Inst_0(15:8)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	XXXX XXXX	Instruction Inst_0(15:8) of ADC miniDSP

Table 106. Page 32 / Register 4: Inst_0(7:0)

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	XXXX XXXX	Instruction Inst_0(7:0) of ADC miniDSP

10.6.6.1 Page 32 / Register 5 Through Page 32 / Register 97

The remaining unreserved registers on page 32 are arranged in groups of three, with each group containing the bits of one instruction. The arrangement is the same as that of registers 2–4 for Instruction 0. Registers 5–7, 8–10, 11–13, ..., 95–97 contain instructions 1, 2, 3, ..., 31, respectively.

Table 107. Page 32 / Register 98 Through Page 32 / Register 127: Reserved

BIT	READ/ WRITE	RESET VALUE	DESCRIPTION
D7–D0	R/W	XXXX XXXX	Reserved. Write only the default value to this register

Copyright © 2008–2015, Texas Instruments Incorporated

10.6.7 Control Registers, Page 33 Through Page 47: ADC DSP Engine Instruction RAM (32:63) Through (480:511)

The structuring of the registers within page 33 through page 43 is identical to that of page 32. Only the instruction numbers differ. The range of instructions within each page is listed in the following table.

PAGE	INSTRUCTIONS
33	32 to 63
34	64 to 95
35	96 to 127
36	128 to 159
37	160 to 191
38	192 to 223
39	224 to 255
40	256 to 287
41	288 to 319
42	320 to 351
43	352 to 383
44	384 to 415
45	416 to 447
46	448 to 479
47	480 to 511

Copyright © 2008–2015, Texas Instruments Incorporated

11 Application and Implementation

NOTE

Information in the following applications sections is not part of the TI component specification, and TI does not warrant its accuracy or completeness. TI's customers are responsible for determining suitability of components for their purposes. Customers should validate and test their design implementation to confirm system functionality.

11.1 Application Information

This typical connection diagram highlights the required external components and system level connections for proper operation of the device in several popular use cases. Each of these configurations can be realized using the Evaluation Modules (EVMs) for the device. These flexible modules allow full evaluation of the device in the most common modes of operation. Any design variation can be supported by TI through schematic and layout reviews. Visit www.e2e.ti.com for design assistance and join the audio amplifier discussion forum for additional information.

11.2 Typical Application

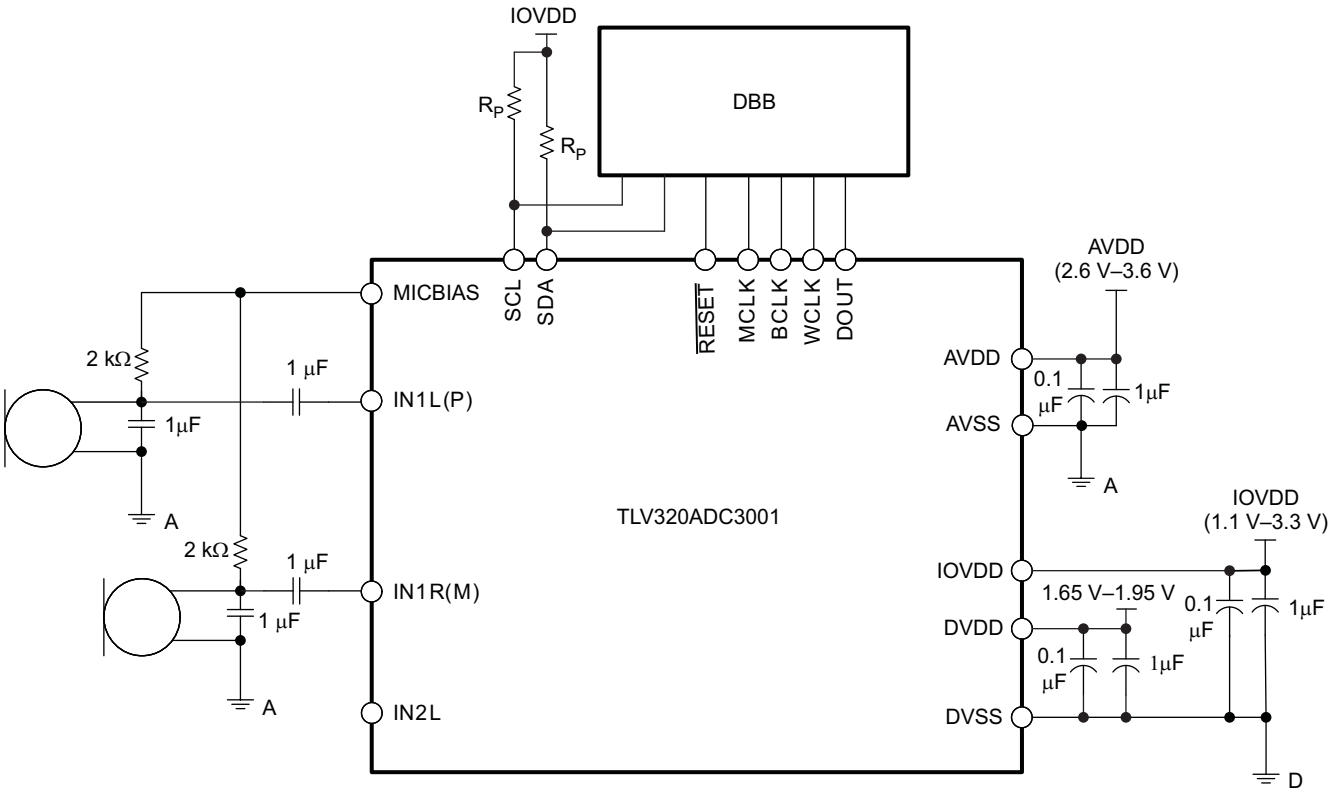


Figure 44. Typical Connections

11.2.1 Design Requirements

Table 108 lists the design parameters for this example.

Table 108. Design Parameters

KEY PARAMETER	SPECIFICATION/UNIT
AVDD	3.3 V
AVDD Supply Current	> 6 mA (PLL on, AGC off, miniDSP off, stereo record, fs = 48 kHz)

Copyright © 2008–2015, Texas Instruments Incorporated

Table 108. Design Parameters (continued)

KEY PARAMETER	SPECIFICATION/UNIT
DVDD	1.8 V
DVDD Supply Current	> 4 mA (PLL on, AGC off, miniDSP off, stereo record, fs = 48 kHz)
IOVDD	1.8 V
Max. MICBIAS Current	4 mA (MICBIAS voltage 2.5 V)

11.2.2 Detailed Design Procedure

11.2.2.1 ADC Setup

The following paragraphs are intended to guide a user through the steps necessary to configure the TLV320ADC3001.

11.2.2.1.1 Step 1

The system clock source (master clock) and the targeted ADC sampling frequency must be identified.

Depending on the targeted performance, the decimation filter type (A, B, or C) and AOSR value can be determined:

- Filter A must be used for 48-kHz high-performance operation; AOSR must be a multiple of 8.
- Filter B must be used for up to 96-kHz operations; AOSR must be a multiple of 4.
- Filter C must be used for up to 192-kHz operations; AOSR must be a multiple of 2.

In all cases, AOSR is limited in its range by the following condition:

$$2.8\ \text{MHz} < \text{AOSR} \times \text{ADC_f}_S < 6.2\ \text{MHz} \quad (6)$$

Based on the identified filter type and the required signal-processing capabilities, the appropriate processing block can be determined from the list of available processing blocks (PRB_R4 to PRB_R18).

Based on the available master clock, the chosen AOSR and the targeted sampling rate, the clock divider values NADC and MADC can be determined. If necessary, the internal PLL can add a large degree of flexibility.

In summary, ADC_CLKIN (derived directly from the system clock source or from the internal PLL) divided by MADC, NADC, and AOSR must be equal to the ADC sampling rate ADC_f_S. The ADC_CLKIN clock signal is shared with the DAC clock-generation block.

$$\text{ADC_CLKIN} = \text{NADC} \times \text{MADC} \times \text{AOSR} \times \text{ADC_f}_S \quad (7)$$

To a large degree, NADC and MADC can be chosen independently in the range of 1 to 128. In general, NADC must be as large as possible as long as the following condition can still be met:

$$\text{MADC} \times \text{AOSR} / 32 \geq \text{RC} \quad (8)$$

RC is a function of the chosen processing block and is listed in the Resource Class column of Table 6.

The common-mode voltage setting of the device is determined by the available analog power supply.

At this point, the following device-specific parameters are known: PRB_Rx, AOSR, NADC, MADC, input and output common-mode values. If the PLL is used, the PLL parameters P, J, D, and R are determined as well.

11.2.2.1.2 Step 2

Setting up the device via register programming:

The following list gives a sequence of items that must be executed in the time between powering the device up and reading data from the device:

1. Define starting point:
   (a) Power up applicable external hardware power supplies
   (b) Set register page to 0
   (c) Initiate SW reset
2. Program clock settings
   (a) Program PLL clock dividers P, J, D, and R (if PLL is used)
   (b) Power up PLL (if PLL is used)

Copyright © 2008–2015, Texas Instruments Incorporated

(c) Program and power up NADC
(d) Program and power up MADC
(e) Program OSR value
(f) Program I^2S word length if required (for example, 20 bits)
(g) Program the processing block to be used

3. Program analog blocks
   (a) Set register page to 1
   (b) Program MICBIAS if applicable
   (c) Program MicPGA
   (d) Program routing of inputs/common mode to ADC input
   (e) Unmute analog PGAs and set analog gain
4. Program ADC
   (a) Set register page to 0
   (b) Power up ADC channel
   (c) Unmute digital volume control and set gain

A detailed example can be found in *Example Register Setup to Record Analog Data Through ADC to Digital Out*.

11.2.2.1.3 Example Register Setup to Record Analog Data Through ADC to Digital Out

A typical EVM I^2C register control script follows to show how to set up the TLV320ADC3001 in record mode with f_S = 44.1 kHz and MCLK = 11.2896 MHz.

```
# Key: w 30 XX YY ==> write to I2C address 0x30, to register 0xXX, data 0xYY
#               # ==> comment delimiter
#
# The following list gives an example sequence of items that must be executed in the time
# between powering the device up and reading data from the device. Note that there are
# other valid sequences depending on which features are used.
#
# ADC3101EVM Key Jumper Settings and Audio Connections:
# 1. Remove Jumpers W12 and W13
# 2. Insert Jumpers W4 and W5
# 3. Insert a 3.5mm stereo audio plug into J9 for
#    single-ended input IN1L(P) - left channel and
#    single-ended input IN1R(M) - right channel
##################################################################
# 1. Define starting point:
#   (a) Power up appicable external hardware power supplies
#   (b) Set register page to 0
#
w 30 00 00
#   (c) Initiate SW Reset
#
w 30 01 01
#
# 2. Program Clock Settings
#   (a) Program PLL clock dividers P,J,D,R (if PLL is necessary)
#
# In EVM, the ADC3001 receives: MCLK = 11.2896 MHz,
# BCLK = 2.8224 MHz, WCLK = 44.1 kHz
#
# Sinve the sample rate is a multiple of the input MCLK then
# no PLL is needed thereby saving power. Use Default (Reset) Settings:
# ADC_CLKIN = MCLK, P=1, R=1, J=4, D=0000
w 30 04 00
w 30 05 11
w 30 06 04
w 30 07 00
w 30 08 00
#
#   (b) Power up PLL (if PLL is necessary) - Not Used in this Example
w 30 05 11
#   (c) Program and power up NADC
#
# NADC = 1, divider powered on
w 30 12 81
#
```


Copyright © 2008–2015, Texas Instruments Incorporated

```
#   (d) Program and power up MADC
#
# MADC = 2, divider powered on
w 30 13 82
#
#   (e) Program OSR value
#
# AOSR = 128 (default)
w 30 14 80
#
#   (f) Program I2S word length as required (16, 20, 24, 32 bits)
#
# mode is i2s, wordlength is 16, slave mode (default)
w 30 1B 00
#
#   (g) Program the processing block to be used
#
# PRB_P1
w 30 3d 01
#
# 3. Program Analog Blocks
#   (a) Set register Page to 1
#
w 30 00 01
#
#   (b) Program MICBIAS if appicable
#
# Not used (default)
w 30 33 00
#
#   (c) Program MicPGA
#
# Left Analog PGA Seeting = 0dB
w 30 3b 00
#
# Right Analog PGA Seeting = 0dB
w 30 3c 00
#
#   (d) Routing of inputs/common mode to ADC input
#   (e) Unmute analog PGAs and set analog gain
#
# Left ADC Input selection for Left PGA = IN1L(P) as Single-Ended
w 30 34 fc
#
# Right ADC Input selection for Right PGA = IN1R(M) as Single-Ended
w 30 37 fc
#
# 4. Program ADC
#
#   (a) Set register Page to 0
#
w 30 00 00
#
#   (b) Power up ADC channel
#
# Power-up Left ADC and Right ADC
w 30 51 c2
#
#   (c) Unmute digital volume control and set gain = 0 dB
#
# UNMUTE
w 30 52 00
#
```

11.2.2.2 MICBIAS

TLV320ADC3001 has a built-in bias voltage output for biasing of microphones. No intentional capacitors must be connected directly to the MICBIAS output for filtering.

Copyright © 2008–2015, Texas Instruments Incorporated

11.2.2.3 Decoupling Capacitors

The TLV320ADC3001 requires adequate power supply decoupling to ensure that the noise and total harmonic distortion (THD) are low. A good ceramic capacitor, typically 0.1 µF, placed as close as possible to the device AVDD, IOVDD and DVDD lead works best. Placing this decoupling capacitor close to the TLV320ADC3001 is important for the performance of the converter. For filtering lower-frequency noise signals, a 1 µF or greater capacitor placed near the device would also help.

11.2.3 Application Curves

Table 109 lists the application curves in the *Typical Characteristics* section.

Table 109. Table of Graphs

GRAPH TITLE	FIGURE
Line Input to ADC FFT Plot	Figure 8
Input-Referred Noise vs. PGA Gain	Figure 9

12 Power Supply Recommendations

The power supplies are designed to operate from 2.6 V to 3.6 V for AVDD, from 1.65 V to 1.95 V for DVDD and from 1.1 V to 3.6 V for IOVDD. Any value out of these ranges must be avoided to ensure the correct behavior of the device. The power supplies must be well regulated. Placing a decoupling capacitor close to the TLV320ADC3001 improves the performance of the device. A low equivalent-series-resistance (ESR) ceramic capacitor with a value of 0.1 µF is a typical choice. If the TLV320ADC3001 is used in highly noise-sensitive circuits, TI recommends to add a small LC filter on the VDD connections.

Copyright © 2008–2015, Texas Instruments Incorporated

13 Layout

13.1 Layout Guidelines

Each system design and PCB layout is unique. The layout must be carefully reviewed in the context of a specific PCB design. However, the following guidelines can optimize the TLV320ADC3001 performance:

The decoupling capacitors for the power supplies must be placed close to the device terminals. Figure 44 shows the recommended decoupling capacitors for the TLV320ADC3001.

For analog differential audio signals, they must be routed differentially on the PCB for better noise immunity. Avoid crossing digital and analog signals to avoid undesirable crosstalk.

Analog and digital grounds must be separated to prevent possible digital noise from affecting the analog performance of the board.

13.2 Layout Example

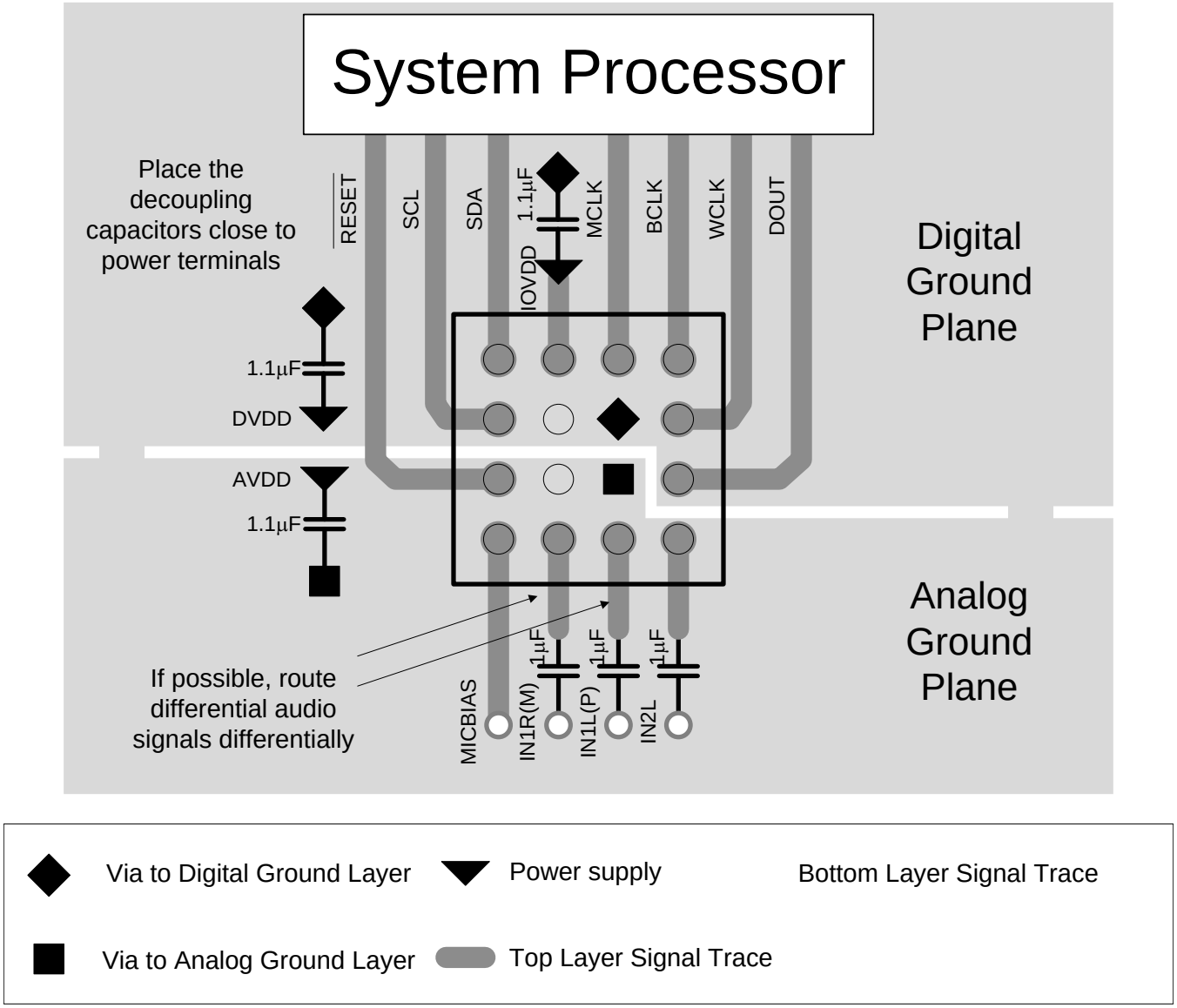


Figure 45. Layout Recommendation

14 Device and Documentation Support

14.1 Community Resources

The following links connect to TI community resources. Linked contents are provided "AS IS" by the respective contributors. They do not constitute TI specifications and do not necessarily reflect TI's views; see TI's Terms of Use.

TI E2E™ Online Community ***TI's Engineer-to-Engineer (E2E) Community.*** Created to foster collaboration among engineers. At e2e.ti.com, you can ask questions, share knowledge, explore ideas and help solve problems with fellow engineers.

E2E Audio Amplifier Forum ***TI's Engineer-to-Engineer (E2E) Community for Audio Amplifiers.*** Created to foster collaboration among engineers. Ask questions and receive answers in real-time.

14.2 Trademarks

NanoFree, PurePath, E2E are trademarks of Texas Instruments.
All other trademarks are the property of their respective owners.

14.3 Electrostatic Discharge Caution



These devices have limited built-in ESD protection. The leads should be shorted together or the device placed in conductive foam during storage or handling to prevent electrostatic damage to the MOS gates.

14.4 Glossary

SLYZ022 — *TI Glossary*.

This glossary lists and explains terms, acronyms, and definitions.

15 Mechanical, Packaging, and Orderable Information

The following pages include mechanical, packaging, and orderable information. This information is the most current data available for the designated devices. This data is subject to change without notice and revision of this document. For browser-based versions of this data sheet, refer to the left-hand navigation.

Copyright © 2008–2015, Texas Instruments Incorporated

PACKAGING INFORMATION

Orderable Device	Status (1)	Package Type	Package Drawing	Pins	Package Qty	Eco Plan (2)	Lead/Ball Finish (6)	MSL Peak Temp (3)	Op Temp (°C)	Device Marking (4/5)	Samples
TLV320ADC3001IYZHR	ACTIVE	DSBGA	YZH	16	3000	Green (RoHS & no Sb/Br)	SNAGCU	Level-1-260C-UNLIM	-40 to 85	ADC3001	Samples
TLV320ADC3001IYZHT	ACTIVE	DSBGA	YZH	16	250	Green (RoHS & no Sb/Br)	SNAGCU	Level-1-260C-UNLIM	-40 to 85	ADC3001	Samples

(1) The marketing status values are defined as follows:
ACTIVE: Product device recommended for new designs.
LIFEBUY: TI has announced that the device will be discontinued, and a lifetime-buy period is in effect.
NRND: Not recommended for new designs. Device is in production to support existing customers, but TI does not recommend using this part in a new design.
PREVIEW: Device has been announced but is not in production. Samples may or may not be available.
OBSOLETE: TI has discontinued the production of the device.

(2) Eco Plan - The planned eco-friendly classification: Pb-Free (RoHS), Pb-Free (RoHS Exempt), or Green (RoHS & no Sb/Br) - please check http://www.ti.com/productcontent for the latest availability information and additional product content details.
TBD: The Pb-Free/Green conversion plan has not been defined.
Pb-Free (RoHS): TI's terms "Lead-Free" or "Pb-Free" mean semiconductor products that are compatible with the current RoHS requirements for all 6 substances, including the requirement that lead not exceed 0.1% by weight in homogeneous materials. Where designed to be soldered at high temperatures, TI Pb-Free products are suitable for use in specified lead-free processes.
Pb-Free (RoHS Exempt): This component has a RoHS exemption for either 1) lead-based flip-chip solder bumps used between the die and package, or 2) lead-based die adhesive used between the die and leadframe. The component is otherwise considered Pb-Free (RoHS compatible) as defined above.
Green (RoHS & no Sb/Br): TI defines "Green" to mean Pb-Free (RoHS compatible), and free of Bromine (Br) and Antimony (Sb) based flame retardants (Br or Sb do not exceed 0.1% by weight in homogeneous material)

(3) MSL, Peak Temp. - The Moisture Sensitivity Level rating according to the JEDEC industry standard classifications, and peak solder temperature.

(4) There may be additional marking, which relates to the logo, the lot trace code information, or the environmental category on the device.

(5) Multiple Device Markings will be inside parentheses. Only one Device Marking contained in parentheses and separated by a "~" will appear on a device. If a line is indented then it is a continuation of the previous line and the two combined represent the entire Device Marking for that device.

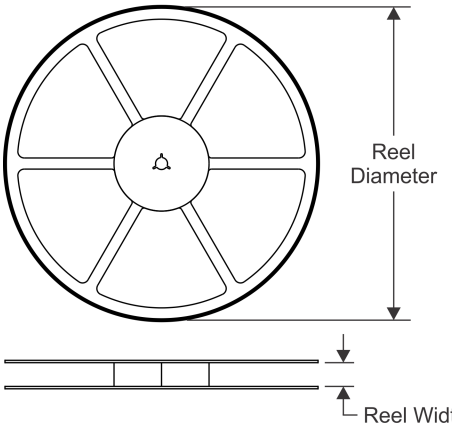
(6) Lead/Ball Finish - Orderable Devices may have multiple material finish options. Finish options are separated by a vertical ruled line. Lead/Ball Finish values may wrap to two lines if the finish value exceeds the maximum column width.

Important Information and Disclaimer: The information provided on this page represents TI's knowledge and belief as of the date that it is provided. TI bases its knowledge and belief on information provided by third parties, and makes no representation or warranty as to the accuracy of such information. Efforts are underway to better integrate information from third parties. TI has taken and continues to take reasonable steps to provide representative and accurate information but may not have conducted destructive testing or chemical analysis on incoming materials and chemicals. TI and TI suppliers consider certain information to be proprietary, and thus CAS numbers and other limited information may not be available for release.

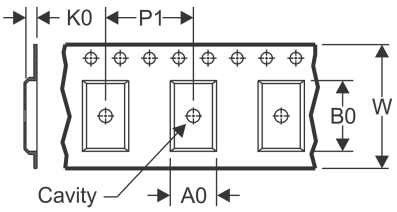
In no event shall TI's liability arising out of such information exceed the total purchase price of the TI part(s) at issue in this document sold by TI to Customer on an annual basis.

TAPE AND REEL INFORMATION

REEL DIMENSIONS

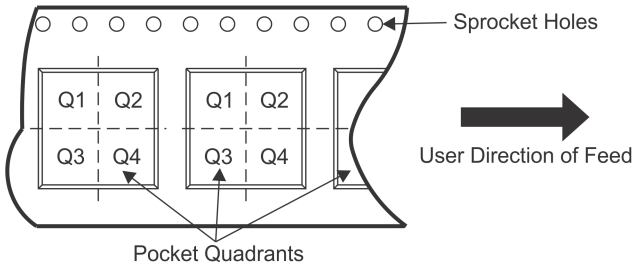


TAPE DIMENSIONS



A0	Dimension designed to accommodate the component width
B0	Dimension designed to accommodate the component length
K0	Dimension designed to accommodate the component thickness
W	Overall width of the carrier tape
P1	Pitch between successive cavity centers

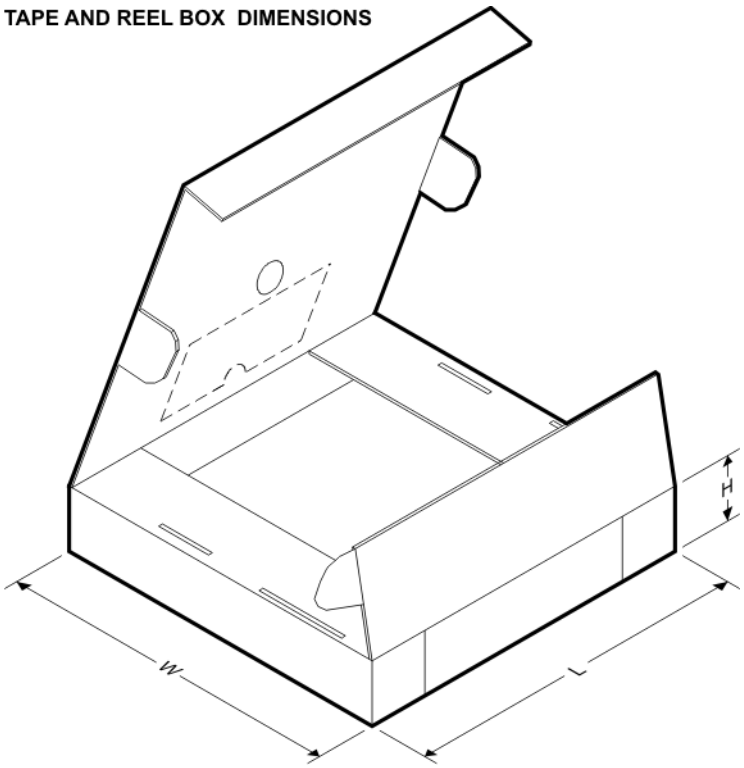
QUADRANT ASSIGNMENTS FOR PIN 1 ORIENTATION IN TAPE



*All dimensions are nominal

Device	Package Type	Package Drawing	Pins	SPQ	Reel Diameter (mm)	Reel Width W1 (mm)	A0 (mm)	B0 (mm)	K0 (mm)	P1 (mm)	W (mm)	Pin1 Quadrant
TLV320ADC3001IYZHR	DSBGA	YZH	16	3000	180.0	8.4	2.38	2.4	0.8	4.0	8.0	Q1
TLV320ADC3001IYZHT	DSBGA	YZH	16	250	180.0	8.4	2.38	2.4	0.8	4.0	8.0	Q1

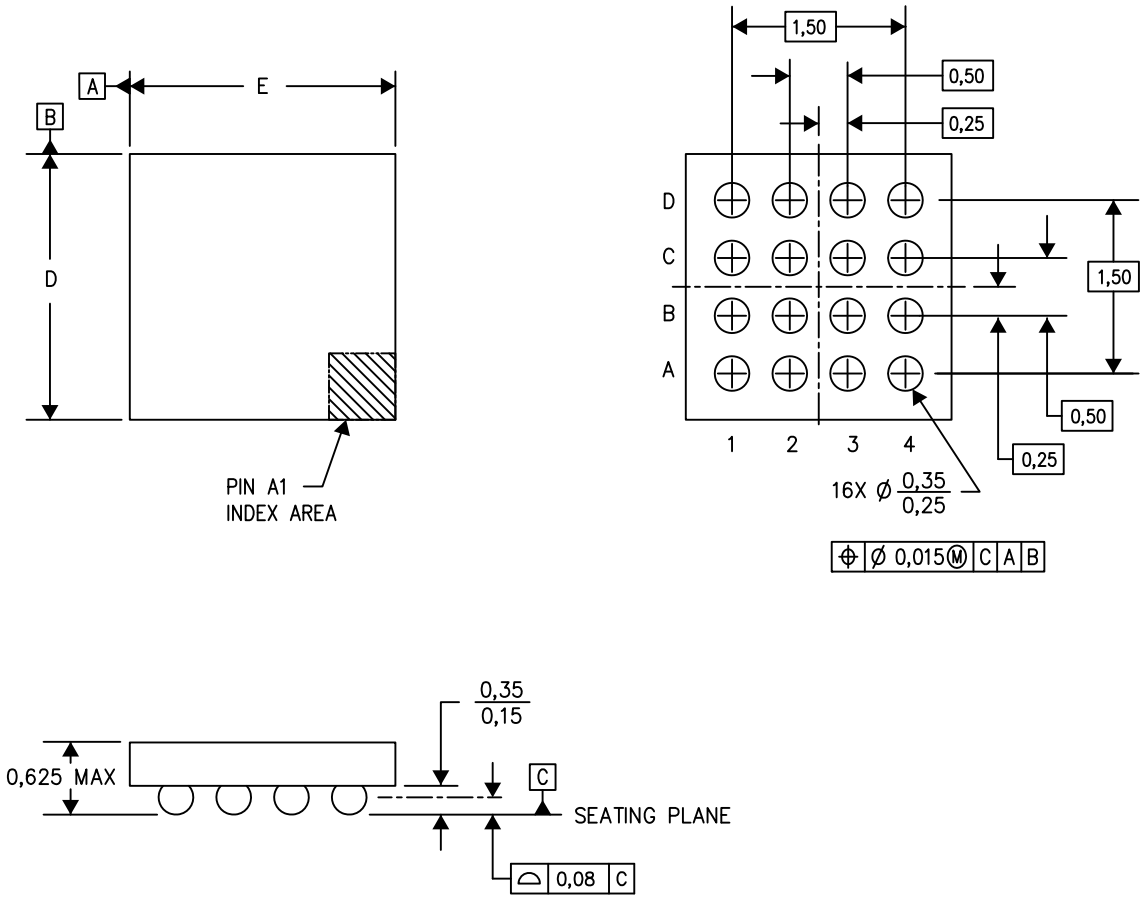
TAPE AND REEL BOX DIMENSIONS



*All dimensions are nominal

Device	Package Type	Package Drawing	Pins	SPQ	Length (mm)	Width (mm)	Height (mm)
TLV320ADC3001IYZHR	DSBGA	YZH	16	3000	182.0	182.0	20.0
TLV320ADC3001IYZHT	DSBGA	YZH	16	250	182.0	182.0	20.0

YZH (S–XBGA–N16) DIE–SIZE BALL GRID ARRAY



D: Max = 2.271 mm, Min = 2.21 mm

E: Max = 2.19 mm, Min = 2.129 mm

4205060/F 05/13

NOTES: A. All linear dimensions are in millimeters.
B. This drawing is subject to change without notice.
C. NanoFree™ package configuration.

NanoFree is a trademark of Texas Instruments.

IMPORTANT NOTICE

Texas Instruments Incorporated and its subsidiaries (TI) reserve the right to make corrections, enhancements, improvements and other changes to its semiconductor products and services per JESD46, latest issue, and to discontinue any product or service per JESD48, latest issue. Buyers should obtain the latest relevant information before placing orders and should verify that such information is current and complete. All semiconductor products (also referred to herein as “components”) are sold subject to TI’s terms and conditions of sale supplied at the time of order acknowledgment.

TI warrants performance of its components to the specifications applicable at the time of sale, in accordance with the warranty in TI’s terms and conditions of sale of semiconductor products. Testing and other quality control techniques are used to the extent TI deems necessary to support this warranty. Except where mandated by applicable law, testing of all parameters of each component is not necessarily performed.

TI assumes no liability for applications assistance or the design of Buyers’ products. Buyers are responsible for their products and applications using TI components. To minimize the risks associated with Buyers’ products and applications, Buyers should provide adequate design and operating safeguards.

TI does not warrant or represent that any license, either express or implied, is granted under any patent right, copyright, mask work right, or other intellectual property right relating to any combination, machine, or process in which TI components or services are used. Information published by TI regarding third-party products or services does not constitute a license to use such products or services or a warranty or endorsement thereof. Use of such information may require a license from a third party under the patents or other intellectual property of the third party, or a license from TI under the patents or other intellectual property of TI.

Reproduction of significant portions of TI information in TI data books or data sheets is permissible only if reproduction is without alteration and is accompanied by all associated warranties, conditions, limitations, and notices. TI is not responsible or liable for such altered documentation. Information of third parties may be subject to additional restrictions.

Resale of TI components or services with statements different from or beyond the parameters stated by TI for that component or service voids all express and any implied warranties for the associated TI component or service and is an unfair and deceptive business practice. TI is not responsible or liable for any such statements.

Buyer acknowledges and agrees that it is solely responsible for compliance with all legal, regulatory and safety-related requirements concerning its products, and any use of TI components in its applications, notwithstanding any applications-related information or support that may be provided by TI. Buyer represents and agrees that it has all the necessary expertise to create and implement safeguards which anticipate dangerous consequences of failures, monitor failures and their consequences, lessen the likelihood of failures that might cause harm and take appropriate remedial actions. Buyer will fully indemnify TI and its representatives against any damages arising out of the use of any TI components in safety-critical applications.

In some cases, TI components may be promoted specifically to facilitate safety-related applications. With such components, TI’s goal is to help enable customers to design and create their own end-product solutions that meet applicable functional safety standards and requirements. Nonetheless, such components are subject to these terms.

No TI components are authorized for use in FDA Class III (or similar life-critical medical equipment) unless authorized officers of the parties have executed a special agreement specifically governing such use.

Only those TI components which TI has specifically designated as military grade or “enhanced plastic” are designed and intended for use in military/aerospace applications or environments. Buyer acknowledges and agrees that any military or aerospace use of TI components which have ***not*** been so designated is solely at the Buyer's risk, and that Buyer is solely responsible for compliance with all legal and regulatory requirements in connection with such use.

TI has specifically designated certain components as meeting ISO/TS16949 requirements, mainly for automotive use. In any case of use of non-designated products, TI will not be responsible for any failure to meet ISO/TS16949.

Products		**Applications**	
Audio	www.ti.com/audio	Automotive and Transportation	www.ti.com/automotive
Amplifiers	amplifier.ti.com	Communications and Telecom	www.ti.com/communications
Data Converters	dataconverter.ti.com	Computers and Peripherals	www.ti.com/computers
DLP® Products	www.dlp.com	Consumer Electronics	www.ti.com/consumer-apps
DSP	dsp.ti.com	Energy and Lighting	www.ti.com/energy
Clocks and Timers	www.ti.com/clocks	Industrial	www.ti.com/industrial
Interface	interface.ti.com	Medical	www.ti.com/medical
Logic	logic.ti.com	Security	www.ti.com/security
Power Mgmt	power.ti.com	Space, Avionics and Defense	www.ti.com/space-avionics-defense
Microcontrollers	microcontroller.ti.com	Video and Imaging	www.ti.com/video
RFID	www.ti-rfid.com		
OMAP Applications Processors	www.ti.com/omap	**TI E2E Community**	e2e.ti.com
Wireless Connectivity	www.ti.com/wirelessconnectivity		

Mailing Address: Texas Instruments, Post Office Box 655303, Dallas, Texas 75265
Copyright © 2015, Texas Instruments Incorporated