











SLES230A - SEPTEMBER 2008-REVISED AUGUST 2015

PCM2912A

# PCM2912A Audio Codec With USB Interface, Mono Microphone Input and Stereo **Headphone Output**

#### **Features**

- On-Chip USB Interface:
  - With Full-Speed Transceivers
  - Fully Compliant With USB 2.0 Specification
  - Certified By USB-IF
  - Partially Programmable Descriptors
  - Adaptive Isochronous Transfer for Playback
  - Asynchronous-Isochronous Transfer for Record
  - **Bus Powered**
- 16-Bit Delta-Sigma ADC and DAC
- Sampling Rate:
  - 8, 11.025, 16, 22.05, 32, 44.1, or 48 kHz
- On-Chip Clock Generator:
  - With Single 6-MHz Clock Source
- Mono ADC with Microphone Input
  - Analog Performance at  $V_{BUS} = 5 \text{ V}$ :

THD+N: 0.01% - SNR: 92 dB

Dynamic Range: 90 dB

**Decimation Digital Filter** 

 Passband Ripple: ±0.05 dB Stop-Band Attenuation: –65 dB

- Single-Ended Voltage Input
- Antialiasing Filter Included
- Digital HPF Included
- Microphone Bias, Microphone Amplifier, and Input PGA
- Stereo DAC With Headphone Output
  - Analog Performance at  $V_{BUS} = 5.0 \text{ V}$ :
    - THD+N: 0.01% ( $R_1 > 10 k\Omega$ )
    - THD+N: 0.02% (R<sub>L</sub> = 32 Ω)
    - SNR: 92 dB
    - Dynamic Range: 90 dB
    - PO: 13 mW (R<sub>L</sub> = 32 Ω)
    - PO: 25 mW (R<sub>I</sub> = 16 Ω)
  - Oversampling Digital Filter
    - Passband Ripple: ±0.1 dB
    - Stop-Band Attenuation: -43 dB
    - Single-Ended Voltage Output
    - Analog LPF Included
    - Sidetone PGA, Output PGA, and HP **Amplifier**

- Multifunctions:
  - Suspend, Playback, and Record Status Flag
  - Microphone Amplifier, Mute, and Gain Control
- Pop/Click Noise-Free
- Single Power-Supply: 5 V Typical (V<sub>BUS</sub>)
- Package: 32-Pin TQFP

# Applications

- **USB** Headset
- **USB** Headphone
- **USB** Speaker
- **USB Featured Consumer Audio Product**
- USB Audio Interface Box
- **USB** Monitor
- Video Conference System

# 3 Description

The PCM2912A is the Texas Instruments single-chip, USB stereo audio codec with a USB, 2.0-compliant, full-speed protocol controller and an analog front-end (AFE) function for headset applications.

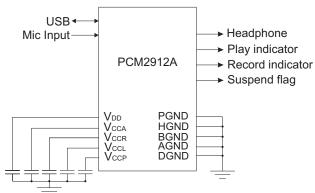
The USB protocol controller works with no software code, but USB descriptors can be modified on The PCM2912A employs SpAct<sup>TM</sup> request. architecture, TI's unique system that recovers the audio clock from USB packet data. On-chip analog PLLs with SpAct enables independent playback and record sampling rates with low clock jitters.

## Device Information<sup>(1)</sup>

PART NUMBER	PACKAGE	BODY SIZE (NOM)
PCM2912A	TQFP (32)	7.00 mm × 7.00 mm

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

# **Block Diagram**





# **Table of Contents**

1	Features 1		9.3 Feature Description	10
2	Applications 1		9.4 Device Functional Modes	19
3	Description 1		9.5 Programming	20
4	Revision History2	10	Application and Implementation	24
5	Device Comparison Table		10.1 Application Information	2
6	Pin Configuration and Functions		10.2 Typical Application	24
7	Specifications	11	Power Supply Recommendations	2
′	7.1 Absolute Maximum Ratings	12	Layout	27
	7.2 ESD Ratings		12.1 Layout Guidelines	27
	7.3 Recommended Operating Conditions		12.2 Layout Example	2
	7.4 Thermal Information	13	Device and Documentation Support	28
	7.5 Electrical Characteristics		13.1 Documentation Support	2
	7.6 Typical Characteristics		13.2 Community Resources	28
8	Parameter Measurement Information		13.3 Trademarks	28
9	Detailed Description		13.4 Electrostatic Discharge Caution	2
9			13.5 Glossary	2
	9.1 Overview       15         9.2 Functional Block Diagram       15	14	Mechanical, Packaging, and Orderable Information	

# 4 Revision History

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

#### Changes from Original (September 2008) to Revision A

**Page** 

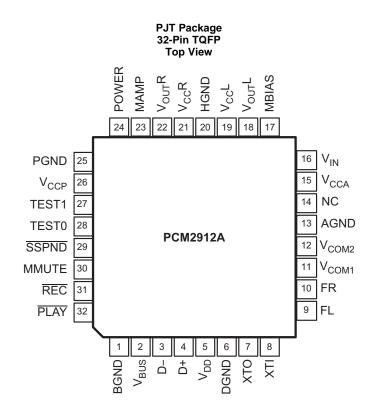
Added Pin Configuration and Functions section, ESD Ratings table, Feature Description section, Device Functional Modes, Application and Implementation section, Power Supply Recommendations section, Layout section, Device and Documentation Support section, Device Comparison table, and Mechanical, Packaging, and Orderable Information section \_\_\_\_\_\_\_1



# 5 Device Comparison Table

FEATURE	PCM2900C	PCM2902C	PCM2903C	PCM2906C	PCM2912A
Supply voltage	5 V (Bus-Powered)	5 V (Bus-Powered)	3.3 V (Bus-Powered)	5 V (Bus-Powered)	5 V (Bus-Powered)
Additional features	_	S/PDIF I/O	S/PDIF I/O	S/PDIF I/O 500-mA Maximum Power Configuration	Mic Bias PGAs Sidetone

# 6 Pin Configuration and Functions



#### **Pin Functions**

I	PIN	1/0	DECODINE	
NAME	NO.	1/0	DESCRIPTION	
AGND	13	_	Analog ground	
BGND	1	_	Reference for internal regulator	
D-	3	I/O	3 differential input/output minus <sup>(1)</sup>	
D+	4	I/O	B differential input/output plus <sup>(1)</sup>	
DGND	6		Digital ground	
FL	9	_	External filter pin of L-channel (optional)	
FR	10	_	External filter pin of R-channel (optional)	
HGND	20	_	Analog ground for headphone amplifier	
MAMP	23	I	Microphone preamplifier gain control (LOW: Preamplifier off, HIGH: Preamplifier on = +20 dB) <sup>(2)</sup>	
MBIAS	17	0	Microphone bias output (0.75 V <sub>CCA</sub> )	
MMUTE	30	I	Microphone mute control, active HIGH (LOW: Mute off, HIGH: Mute on) (3)	
NC	14	_	Not connected	

- (1) LV-TTL level
- (2) 3.3-V CMOS level input.
- (3) 3.3-V CMOS level input with internal pulldown resistor.



# Pin Functions (continued)

	PIN		DECODINE
NAME	NO.	1/0	DESCRIPTION
PGND	25	_	Analog ground for microphone bias, microphone amplifier, and PGA
PLAY	32	0	Status output for playback (LOW: Playback, FLASH: Mute on playback, HIGH: Stop) (4)
POWER	24	I	Power consumption declaration select pin (LOW: 100 mA, HIGH: 500 mA) (2)
REC	31	0	Status output for record (LOW: Record, FLASH: Mute on recode, HIGH: Stop) (4)
SSPND	29	0	Suspend flag (LOW: Suspend, HIGH: Operational state)
TEST0	28	I	Test pin. Must be set to LOW <sup>(2)</sup>
TEST1	27	I	Test pin. Must be set to HIGH <sup>(2)</sup>
V <sub>BUS</sub>	2	_	Connect to USB power (V <sub>BUS</sub> )
$V_{CCA}$	15	_	Analog power supply
V <sub>CC</sub> L	19	_	Analog power supply for headphone amplifier of L-channel (5)
$V_{CCP}$	26	_	Analog power supply for PLL <sup>(5)</sup>
$V_{CC}R$	21	_	Analog power supply for headphone amplifier of R-channel (5)
V <sub>COM1</sub>	11	_	Common voltage for ADC, DAC, and analog front-end ( $V_{\text{CCA}}/2$ ). Decoupling capacitor must be connected to AGND.
$V_{COM2}$	12	_	Common voltage for headphone (V <sub>CCA</sub> /2). Decoupling capacitor must be connected to AGND.
$V_{DD}$	5	_	Digital power supply <sup>(5)</sup>
$V_{IN}$	16	I	ADC microphone input
$V_{OUT}L$	18	0	Headphone output for L-channel
V <sub>OUT</sub> R	22	0	Headphone output for R-channel
XTI	8	I	Crystal oscillator input <sup>(2)</sup>
XTO	7	0	Crystal oscillator output

<sup>(4) 5-</sup>V tolerant, open-drain.(5) Connect decoupling capacitor to corresponding ground.



# 7 Specifications

# 7.1 Absolute Maximum Ratings

Over operating free-air temperature range (unless otherwise noted). (1)(2)

		MIN	MAX	UNIT
Supply voltage	V <sub>BUS</sub>	-0.3	6.5	V
Ground voltage differences	BGND, PGND, AGND, HGND, DGND	-0.1	0.1	V
Input voltage	V <sub>CCP</sub> , V <sub>CCA</sub> , V <sub>CCL</sub> , V <sub>CCR</sub> , V <sub>DD</sub>	-0.3	4	V
Digital input voltage	PLAY, REC	-0.3	6.5	V
	D+, D-, XTI, XTO, MMUTE, TEST0, TEST1, POWER, MAMP, SSPND	-0.3	4	V
Analog input voltage	MBIAS, V <sub>IN</sub> , V <sub>COM1</sub> , V <sub>COM2</sub> , V <sub>OUT</sub> L, V <sub>OUT</sub> R, FR, FL	-0.3	4	V
Input current (any pins	except supplies)	-10	10	mA
Ambient temperature u	Ambient temperature under bias		125	
Junction temperature			150	°C
Storage temperature, T	- stg	-55	150	

<sup>(1)</sup> 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.

# 7.2 ESD Ratings

			VALUE	UNIT
.,	Electrostatic	Human body model (HBM), per ANSI/ESDA/JEDEC JS-001 (1)	±3000	V
V <sub>(ESD)</sub>	discharge	Charged device model (CDM), per JEDEC specification JESD22-C101 <sup>(2)</sup>	±1000	V

<sup>(1)</sup> JEDEC document JEP155 states that 500-V HBM allows safe manufacturing with a standard ESD control process.

# 7.3 Recommended Operating Conditions

Over operating free-air temperature range (unless otherwise noted).

		MIN	NOM	MAX	UNIT
V <sub>BUS</sub>	Supply voltage	4.35	5	5.25	V
	Analog input voltage, full scale (-0 dB)		0.43 V <sub>CCA</sub>		$V_{PP}$
	Digital input logic family		TTL		
	Digital input clock frequency	5.997	6	6.003	MHz
	Analog output load resistance	32			Ω
	Analog output load capacitance			100	pF
	Digital output load capacitance			10	pF
T <sub>A</sub>	Operating free-air temperature	-25		70	°C

#### 7.4 Thermal Information

7.7 1110			
		PCM2912A	
	THERMAL METRIC <sup>(1)</sup>	PJT (TQFP)	UNIT
		32 PINS	
$R_{\theta JA}$	Junction-to-ambient thermal resistance	69.6	°C/W
$R_{\theta JC(top)}$	Junction-to-case (top) thermal resistance	10.6	°C/W
$R_{\theta JB}$	Junction-to-board thermal resistance	34.6	°C/W
ΨЈТ	Junction-to-top characterization parameter	0.3	°C/W
$\Psi_{JB}$	Junction-to-board characterization parameter	33.9	°C/W

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

<sup>(2)</sup> All voltage values are with respect to network ground terminal.

<sup>(2)</sup> JEDEC document JEP157 states that 250-V CDM allows safe manufacturing with a standard ESD control process.



#### 7.5 Electrical Characteristics

All specifications at  $T_A$  = 25°C,  $V_{BUS}$  = 5 V,  $f_S$  = 44.1 kHz,  $f_{IN}$  = 1 kHz, and 16-bit data, unless otherwise noted. for Host interface, apply USB revision 2.0, full-speed. For audio data format, use UsB isochronous data format.

	PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
INPUT LO	OGIC		<u>'</u>			
V <sub>IH</sub>	Input logic high level		2		3.3	1/00
V <sub>IL</sub>	Input logic low level				0.8	VDC
_	Input logic high current <sup>(1)</sup> (2)	V <sub>IN</sub> = 3.3 V	-10		10	
I <sub>IH</sub>	Input logic high current <sup>(3)</sup>	V <sub>IN</sub> = 3.3 V		65	100	μΑ
	Input logic low current <sup>(1)</sup> (2)	V <sub>IN</sub> = 0 V	-10		10	
I <sub>IL</sub>	Input logic low current <sup>(3)</sup>	V <sub>IN</sub> = 0 V	-10		10	μΑ
OUTPUT	LOGIC					
.,	Output logic high level (1)	I <sub>OH</sub> = -10 mA	2.9			1/00
$V_{OH}$	Output logic high level (4)	$I_{OH} = -2 \text{ mA}$	2.8			VDC
	Output logic low level <sup>(1)</sup>	I <sub>OL</sub> = 10 mA			0.3	
$V_{OL}$	Output logic low level (4)	I <sub>OL</sub> = 2 mA			0.5	VDC
	Output logic low level (5)	I <sub>OL</sub> = 8 mA			0.5	
I <sub>OH</sub>	Output leak current <sup>(5)</sup>	V <sub>IN</sub> = 5 V			±10	μA
CLOCK F	REQUENCY					
	Input clock frequency, XTI		5.997	6.000	6.003	MHz
MICROPH	HONE BIAS					
	Output voltage			0.75 V <sub>CCA</sub>		VDC
	Output current			2		mA
	Output noise	$R_L = 1 k\Omega$		5		$\mu V_{RMS}$
ANALOG	-TO-DIGITAL CONVERTER (ADC) CHA	RACTERISTICS				
	Resolution			16		Bits
	Audio data channel			1		Channel
	Sampling frequency			8 11.025 16 22.05 32 44.1 48		kHz
ADC DYN	NAMIC PERFORMANCE <sup>(6)</sup>					
THD+N	Total harmonic distortion plus noise	$V_{IN} = -1$ dB of 0.43 $V_{CCA}$		0.01%	0.02%	
	Dynamic range	A-weighted	82	90		dB
SNR	Signal-to-noise ratio	A-weighted	84	92		dB
ADC DC	ACCURACY					
	Gain error			±2	±10	% of FSR
	Bipolar zero error			±0		% of FSR
ANALOG	INPUT		<u>'</u>			
	Input voltage			0.43 V <sub>CCA</sub>		V <sub>PP</sub>
	Center voltage			0.5 V <sub>CCA</sub>		V
		-3 dB		150		kHz
	Antialiasing filter frequency response					

<sup>(1)</sup> Pins 3, 4: D-, D+.

Submit Documentation Feedback

Copyright © 2008-2015, Texas Instruments Incorporated

<sup>(2)</sup> Pins 8, 23, 24, 27, 28: XTI, MAMP, POWER, TEST1, TEST0

<sup>(3)</sup> Pin 30: MMUTE

Pins 7, 29: XTO, <u>SSPND</u> Pins 31, 32: <u>REC</u>, <u>PLAY</u>.

<sup>(5)</sup> 

f<sub>IN</sub> = 1 kHz, using Audio Precision™ System Two™, RMS mode with 20-kHz LPF, 400-Hz HPF in calculation. Microphone amplifier = 0 dB, PGA = 0 dB.



# **Electrical Characteristics (continued)**

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted. for Host interface, apply USB revision 2.0, full-speed. For audio data format, use UsB isochronous data format.

	PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
MICROPH	ONE AMPLIFIER					
	Gain		0		20	dB
	Input impedance			20		kΩ
INPUT PO	GA .					
	Gain range		-12		30	dB
	Gain step size			1		dB
ADC DIGI	ITAL FILTER PERFORMANCE				·	
	Passband				0.454 f <sub>S</sub>	Hz
	Stop band		0.583 f <sub>S</sub>			Hz
	Passband ripple		-0.02		0.02	dB
	Stop-band attenuation		-65			dB
	Delay time			17.4/f <sub>S</sub>		S
	HPF frequency response	-3 dB	(	0.078 f <sub>S</sub> /1000		Hz
DIGITAL-	TO-ANALOG CONVERTER (DAC) CHA	ARACTERISTICS				
	Resolution			16		Bits
	Audio data channel			1, 2		Channel
	Sampling frequency			8 11.025 16 22.05 32 44.1 48		kHz
DAC DYN	IAMIC PERFORMANCE <sup>(7)</sup>	1	T			
THD+N	Total harmonic distortion plus noise	$R_L > 10 \text{ k}\Omega$ , $V_{OUT} = 0 \text{ dB of } 0.6 \text{ V}_{CCA}$		0.01%	0.02%	
		$R_L = 32 \Omega$ , $V_{OUT} = 0 dB of 0.55 V_{CCA}$		0.02%	0.05%	
	Dynamic range	EIAJ, A-weighted	82	90		dB
SNR	Signal-to-noise ratio	EIAJ, A-weighted	84	92		dB
	Channel separation	$R_L > 10 \text{ k}\Omega$	80	88		dB
DAC DC	ACCURACY	1	T			
	Gain mismatch channel-to-channel		-10	±2	10	% of FSR
	Gain error		-10	±2	10	% of FSR
	Bipolar zero error			±3		% of FSR
ANALOG	OUTPUT	1	T			
	Output voltage	$R_L > 10 \text{ k}\Omega$		0.6 V <sub>CCA</sub>		$V_{PP}$
		$R_L = 32 \Omega$		0.55 V <sub>CCA</sub>		
	Center voltage			0.5 V <sub>CCA</sub>		V
	Output power	R <sub>L</sub> = 32 Ω		13		mW
	pa- paa-	R <sub>L</sub> = 16 Ω		25		
	Load impedance (AC coupling)	Line	10			kΩ
	(	Headphone	16	32		Ω
	LPF frequency response	–3 dB		140		kHz
	- 1:	f = 20 kHz		-0.1		dB

<sup>(7)</sup> f<sub>OUT</sub> = 1 kHz, using Audio Precision System Two, RMS mode with 20-kHz LPF, 400-Hz HPF. Output attenuator = 0 dB, Sidetone = Mute.



# **Electrical Characteristics (continued)**

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted. for Host interface, apply USB revision 2.0, full-speed. For audio data format, use UsB isochronous data format.

	PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
SIDETONE	PROGRAMMABLE ATTENUATOR					
	Gain range		-76		0	dB
	Gain step size			1		dB
OUTPUT P	ROGRAMMABLE ATTENUATOR					
	Gain range		-76		0	dB
	Gain step size			1		dB
ANALOG I	OOPBACK PERFORMANCE (8)					
TUD N		$R_L > 10 \text{ k}\Omega$ , $V_{IN} = 0 \text{ dB of } 0.43 \text{ V}_{CCA}$		0.01%	0.02%	
THD+N	Total harmonic distortion plus noise	$R_L = 32 \Omega$ , $V_{IN} = 0 dB of 0.43 V_{CCA}$		0.02%	0.05%	
	Dynamic range	EIAJ, A-weighted	82	90		dB
SNR	Signal-to-noise ratio	EIAJ, A-weighted	84	92		dB
DAC DIGIT	AL FILTER PERFORMANCE					
	Passband				0.445 f <sub>S</sub>	Hz
	Stop band		0.555 f <sub>S</sub>			Hz
	Passband ripple				±0.1	dB
	Stop-band attenuation		-43			dB
	Delay time			14.3/f <sub>S</sub>		S
POWER-S	UPPLY REQUIREMENTS		"		1	
V <sub>BUS</sub>	Voltage range	Bus-powered	4.35	5	5.25	VDC
	0 1	ADC, DAC operation ( $R_L = 32 \Omega$ )		85	100	mA
	Supply current	Suspend mode (9)		220	300	μA
		ADC, DAC Operation		425	500	mW
	Power dissipation	Suspend mode <sup>(9)</sup>		0.8	1	mW
$V_{\rm CCP}, V_{\rm CCL}$ $V_{\rm CCR}, V_{\rm CCA}$ $V_{\rm DD}$	Internally-generated power supply voltage (10)		3	3.3	3.6	VDC
	TURE RANGE	·	1		1	
	Operation temperature		-25		85	°C

<sup>(8)</sup> MIC Amp = 0 dB, Sidetone attenuator = 0 dB, Output attenuator = 0 dB.

Submit Documentation Feedback

Copyright © 2008–2015, Texas Instruments Incorporated

<sup>(9)</sup> Under USB suspend state

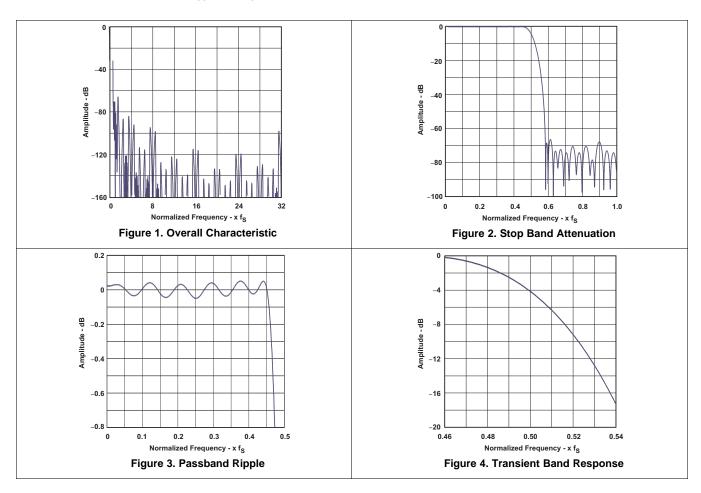
<sup>(10)</sup> Pins 5, 15, 19, 21, 26: V<sub>DD</sub>, V<sub>CCA</sub>, V<sub>CCL</sub>, V<sub>CCR</sub>, V<sub>CCP</sub>.



# 7.6 Typical Characteristics

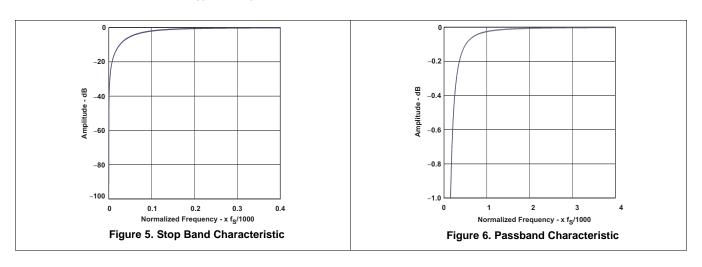
# 7.6.1 ADC Digital Decimation Filter Frequency Response

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



# 7.6.2 ADC Digital High-Pass Filter Frequency Response

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.

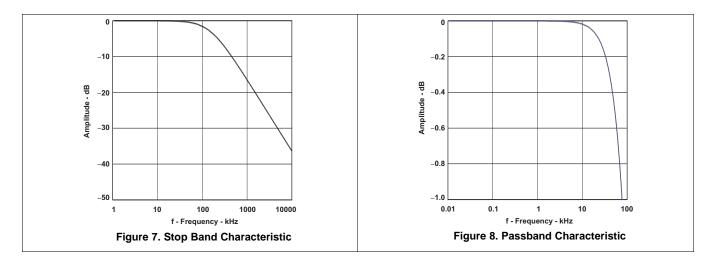


Copyright © 2008–2015, Texas Instruments Incorporated



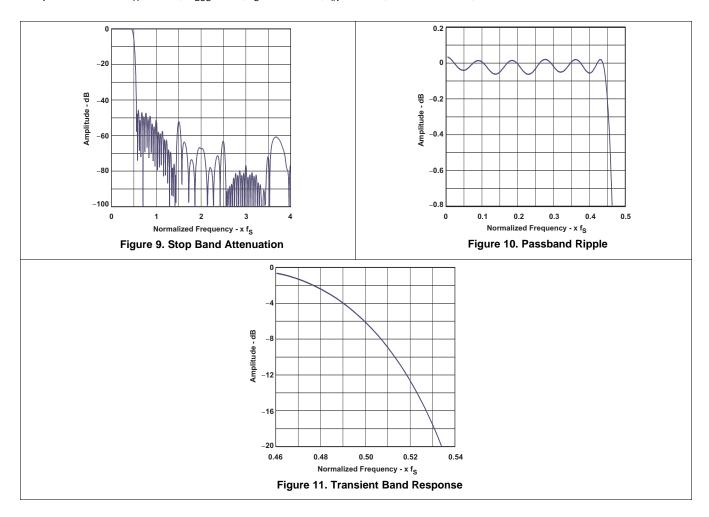
# 7.6.3 ADC Analog Antialiasing Filter Frequency Response

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



## 7.6.4 DAC Digital Interpolation Filter Frequency Response

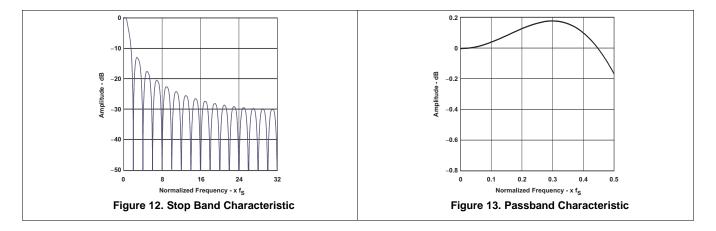
All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.





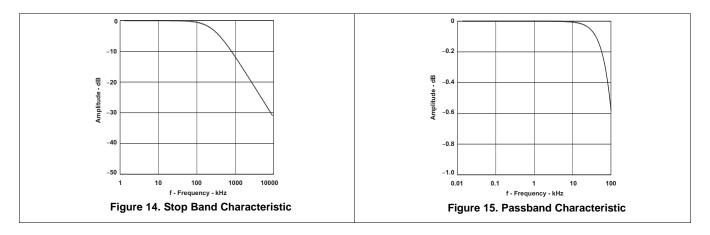
# 7.6.5 DAC Analog FIR Filter Frequency Response

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



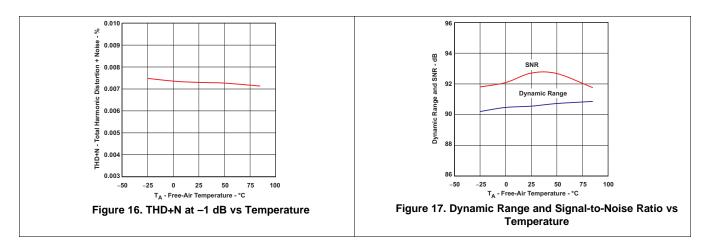
#### 7.6.6 DAC Analog Low-Pass Filter Frequency Response

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



#### 7.6.7 ADC

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.

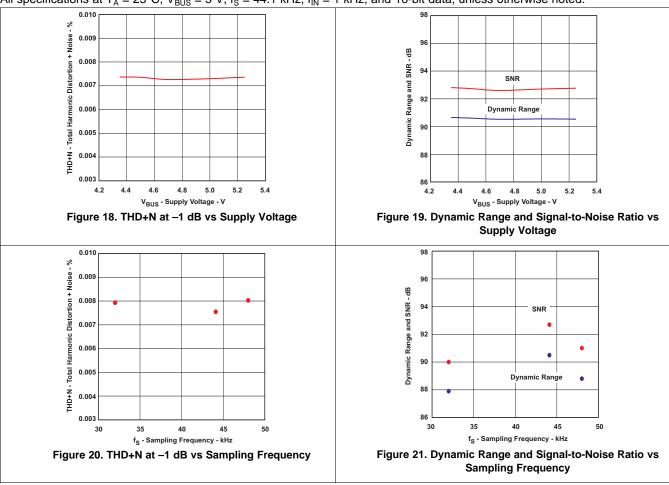


Copyright © 2008–2015, Texas Instruments Incorporated

# TEXAS INSTRUMENTS

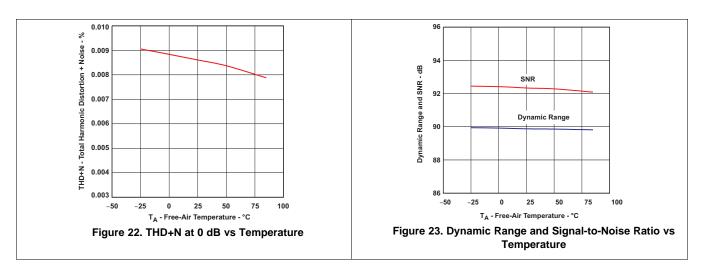
# **ADC** (continued)

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



#### 7.6.8 DAC

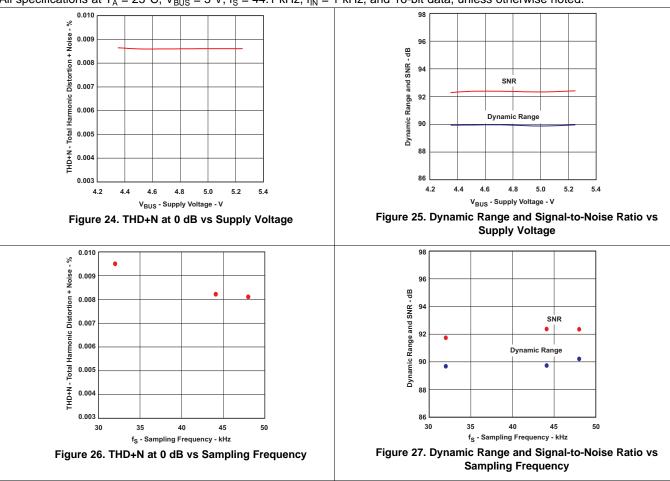
All specifications at  $T_A = 25^{\circ}C$ ,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.





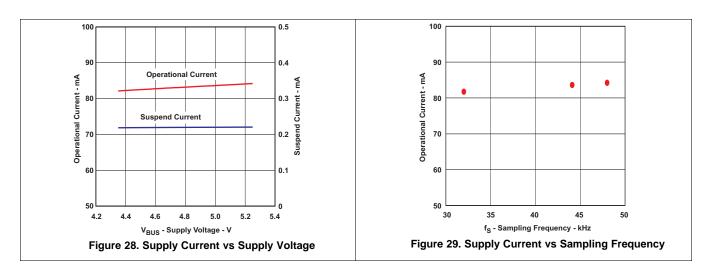
# **DAC** (continued)

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



## 7.6.9 Supply Current

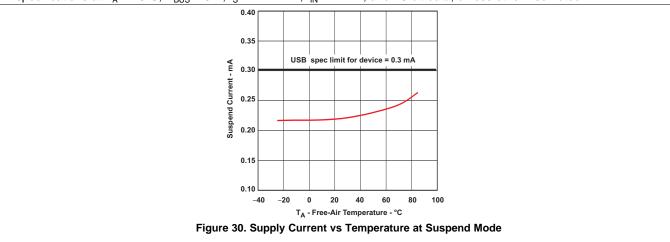
All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



# TEXAS INSTRUMENTS

# **Supply Current (continued)**

All specifications at  $T_A = 25$ °C,  $V_{BUS} = 5$  V,  $f_S = 44.1$  kHz,  $f_{IN} = 1$  kHz, and 16-bit data, unless otherwise noted.



# 8 Parameter Measurement Information

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

Submit Documentation Feedback

Copyright © 2008–2015, Texas Instruments Incorporated



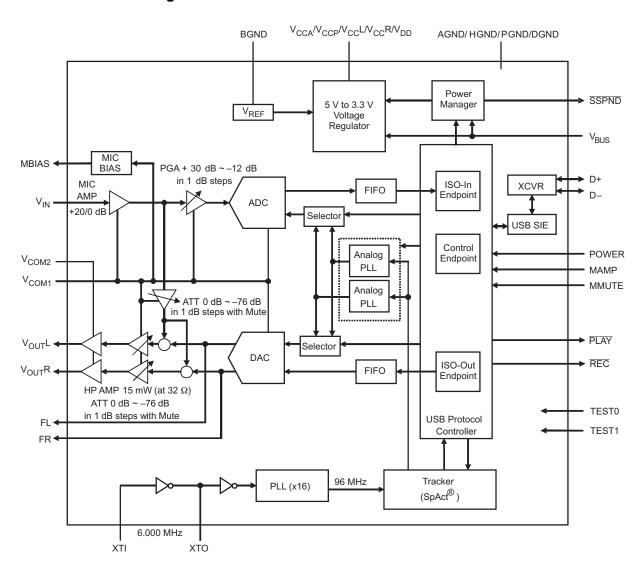
## 9 Detailed Description

9.1概述PCM2912A是一个具有USB连接能力的音频编解码器和一个用于耳机应用的模拟前端。 PCM2912A是一种使用USB电压源的总线供电设备。该PCM2912A符合USB2.0标准连接要求。该设备具有模拟和数字输入和输出;具有用于输入和输出数据的数字1SB接口;模拟输入直接路由到A/D转换器和模拟输出。 麦克风输入有一个可选的麦克风放大器,固定20d8。该PCM2912A有完整分解出标志,可用作LED指示器的回放和记录。 PCM2912A需要12ML2的时钟;它可以由外部时钟提供,也可以由内置的晶体谐振器产生。

The PCM2912A is an audio codec with USB connection capability and an analog front-end for headset applications.

The PCM2912A is a bus-powered device that uses the USB voltage source. The PCM2912A meets the requirements of USB2.0 standard connection. This device has analog and digital inputs and outputs; it has a digital USB interface for input and output data; the analog input is directly routed to the A/D converter and to the analog output. The microphone input has an optional mic amp with fixed +20 dB. The PCM2912A has 2 digital output flags that can be used as LED indicators for Playback and Record. The PCM2912A requires a 12-MHz clock; it can be provided by an external clock or generated by a built-in crystal resonator.

#### 9.2 Functional Block Diagram



Product Folder Links: PCM2912A



#### 9.3 Feature Description

9.3.1 Clock and Reset

该PCM2912A需要一个6MHz( $\pm$ 500p pm)时钟,用于USB功能和音频功能,该时钟可以从内置的具有6MHz晶体谐振器的晶体振荡器产生。 所述6MHz晶体谐振器必须与具有一个高(1-M) 电阻和两个小电容的XT (引脚9)和XTO(引脚7)连接,其电容取决于晶体谐振器的负载电容。 外部时钟可以通过XTI提供;如果提供外部时钟,则必须保持XTO打开。 由于没有时钟禁用信号,不建议使用外部时钟供应。 SSPND(引脚29) 无法使用时钟禁用。 该FCM2912A具有内部电源复位电路,当VBUS(引脚2)超过2.5V,典型(2.2V至2.7V)时自动工作,在内部复位释放之前,大约需要700 $\mu$ s。

The PCM2912A requires a 6-MHz ( $\pm 500$  ppm) clock for USB function and audio function, which can be generated from a built-in crystal oscillator with a 6-MHz crystal resonator. The 6-MHz crystal resonator must be connected to XTI (pin 8) and XTO (pin 7) with one high (1-M $\Omega$ ) resistor and two small capacitors, whose capacitance depends on the load capacitance of the crystal resonator. An external clock can be supplied through XTI; if an external clock is supplied, XTO must be left open. Because there is no clock disabling signal, using the external clock supply is not recommended.  $\overline{\text{SSPND}}$  (pin 29) is unable to use clock disabling.

The PCM2912A has an internal power-on-reset circuit, which works automatically when  $V_{BUS}$  (pin 2) exceeds 2.5 V, typical (2.2 V to 2.7 V), and approximately 700  $\mu$ s is required until the internal reset is released.

#### 9.3.2 DAC

The PCM2912A has a stereo delta-sigma DAC that uses a 64-f<sub>S</sub> oversampling technique with an 8-f<sub>S</sub> oversampling digital filter. DAC outputs are provided through the headphone amplifier;  $V_{OUT}L$  (pin 18) and  $V_{OUT}R$  (pin 22) provide 13 mW at 32  $\Omega$  and 0.6  $V_{CC}L/V_{CC}R$   $V_{PP}$  at a 10-k $\Omega$  load.

#### 9.3.3 ADC

The PCM2912A has a mono delta-sigma ADC that uses a  $64-f_S$  oversampling technique with a  $1/64-f_S$  decimation digital filter. The microphone input,  $V_{IN}$  (pin 16), is fed to the ADC through a +20-dB microphone amplifier and the PGA, which has +30 dB to -12 dB in 1-dB steps.

#### 9.3.4 Microphone Bias

The PCM2912A has a microphone bias generator, which provides a low-noise, 0.75-V<sub>CCA</sub>, 2-mA source current output with appropriate output impedance for electret-microphone driving. This output, MBIAS (pin 17), should be bypassed to AGND (pin 13) through an appropriate capacitor to reduce the output noise level.

#### 9.3.5 Microphone Amplifier

The PCM2912A has a low-noise, single-ended, mono microphone amplifier with a mute function that is controlled by MUTE (pin 30). The signal gain is selectable by MAMP (pin 23). The noise level at the input node is 5  $\mu$ V<sub>RMS</sub>, and the input impedance is 20 k $\Omega$ .

#### 9.3.6 Input PGA

The PCM2912A also has a low-noise input, programmable gain amplifier (PGA) for the microphone amplifier output/ADC input, with a gain range of +30 dB to -12 dB in 1dB/step.

#### 9.3.7 Sidetone Programmable Attenuator

The PCM2912A has a low-noise, sidetone programmable attenuator with a mute function for the sidetone signal path (microphone amplifier output to output PGA input), and a gain range of 0 dB to -76 dB in 1 dB/step.

#### 9.3.8 Output Programmable Attenuator

The PCM2912A has a low-noise output programmable attenuator with a mute function for mixed signal, which affects DAC output signal and sidetone signal. The output PGA gain range is 0 dB to -76 dB in 1 dB/step.

#### 9.3.9 $V_{COM1}$ and $V_{CCM2}$

 $V_{COM2}$  (pin 12) is provided for the center voltage of the headphone amplifier.  $V_{COM1}$  (pin 11) is provided for the center voltage of all other analog circuits. Each  $V_{COM}$  pin must be decoupled with an appropriate capacitor. Because the headphone output is disconnected when entering the suspend state, determining the capacitance is important to prevent pop noise, especially for  $V_{COM2}$  (pin 12). The equivalent resistance of  $V_{COM2}$  is 500 k $\Omega$ , and  $V_{COM1}$  is 15 k $\Omega$ .



# **Feature Description (continued)**

#### 9.3.10 Filter Pins

FL (pin 9) and FR (pin 10) are provided to make a low-pass filter (LPF) to decrease the DAC outband noise, as shown in Figure 31. This filter is optional.

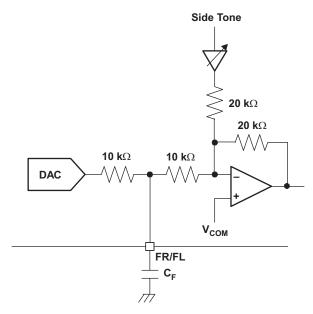


Figure 31. Filter Circuit

#### 9.3.11 Interface Sequence

#### 9.3.11.1 Power-On, Attach, and Play Back Sequence

The PCM2912A is ready for setup when the reset sequence has finished and the USB bus is attached. After a connection has been established, the PCM2912A is ready to accept USB audio data. While waiting for the audio data (that is, in an idle state), the analog output is set to bipolar zero (BPZ).

When receiving the audio data, the PCM2912A stores the first audio packet, which contains 1-ms audio data, into the internal storage buffer. The PCM2912A starts playing the audio data when the subsequent Start of Frame (SOF) packet is detected, as shown in Figure 32.



# **Feature Description (continued)**

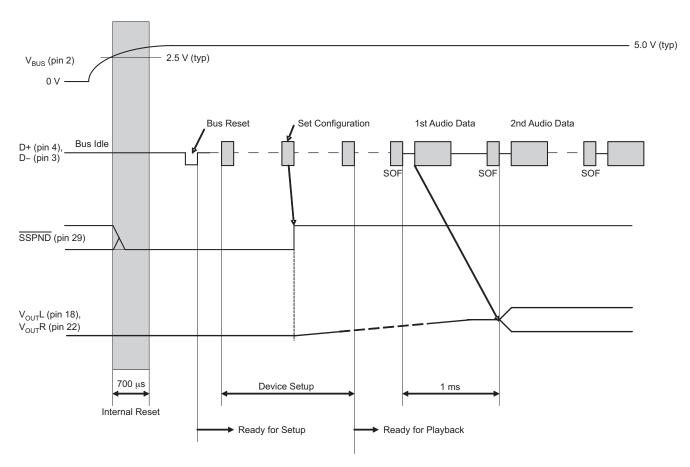


Figure 32. Initial Sequence

#### 9.3.11.2 Play, Stop, and Detach Sequence

When the host finishes or aborts the play back process, the PCM2912A stops playing after last audio data has played, as shown in Figure 33.

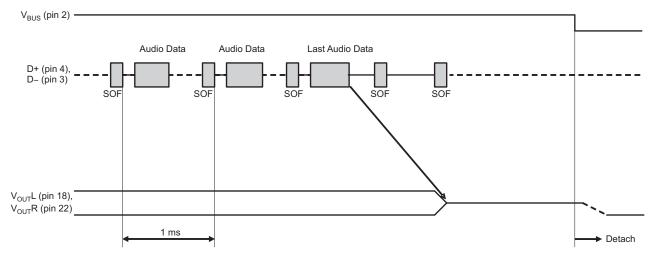


Figure 33. Play, Stop, and Detach



## **Feature Description (continued)**

#### 9.3.11.3 Record Sequence

Figure 34 illustrates how the PCM2912A records the audio into the internal memory after receiving the SET\_INTERFACE command.

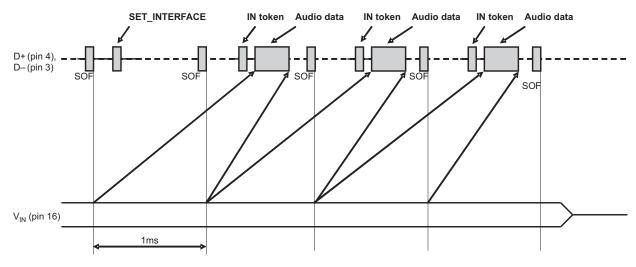


Figure 34. Record Sequence

# 9.3.11.4 Suspend and Resume Sequence

The PCM2912A enters a suspend state when it sees a constant idle state on the USB bus after approximately 5 ms. When the PCM2912A enters the suspend state, the SSPND flag (pin 29) is asserted. The PCM2912A wakes up immediately after detecting the non-idle state on the USB bus. Figure 35 illustrates these actions.

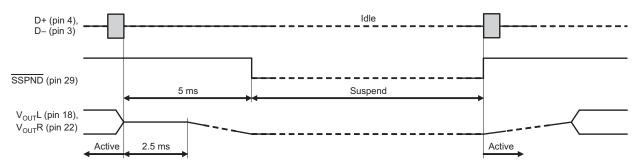


Figure 35. Suspend and Resume

#### 9.4 Device Functional Modes

The PCM2912A is a USB-controlled device. The PCM2912A is a codec, with an analog input (that goes to an A/D converter) and analog output (that comes from a D/A converter), alongside the analog path that goes from the microphone input to the headphone output. A wider explanation of these operational modes is shown in *Feature Description*.

The PCM2912A has hardware controls to turn on and off the integrated microphone preamplifier and the microphone input itself (MAMP and MMUTE, respectively); the microphone preamplifier has +20-dB fixed gain.



# 9.5 Programming

#### 9.5.1 USB Interface

Control data and audio data are transferred to the PCM2912A via D+ (pin 4) and D- (pin 3). All data transferred to/from the PCM2912A are performed at full speed. Table 1 summarizes the device descriptor. The device descriptor can be modified on request.

**Table 1. Device Descriptor** 

DEVICE DESCRIPTOR	DESCRIPTION
USB revision	2.0 compliant
Device class	0x00 (device defined in interface level)
Device sub class	0x00 (not specified)
Device protocol	0x00 (not specified)
Maximum packet size for endpoint 0	8-byte
Vendor ID	0x08BB
Product ID	0x2912
Device release number	0x0100 (1.00)
Number of configurations	1
Vendor string	String #1 (refer to Table 3)
Product string	String #2 (refer to Table 3)
Serial number	Not supported

Table 2 lists the configuration descriptor. The configuration descriptor can be modified on request.

**Table 2. Configuration Descriptor** 

CONFIGURATION DESCRIPTOR	DESCRIPTION
Interface	Three interfaces
Power attribute	0x80 (Bus powered, no remote wakeup)
Max power	0x32 (100 mA at POWER = Low) / 0xFA (500mA at POWER = High)

Table 3 summarizes the string descriptor. The string descriptor can be modified on request.

**Table 3. String Descriptor** 

STRING DESCRIPTOR	DESCRIPTION
#0	0x0409
#1	Burr-Brown from TI
#2	USB audio CODEC



#### 9.5.1.1 Device Configuration

Figure 36 illustrates the USB audio function topology. The PCM2912A has three interfaces. Each interface is constructed with some alternative settings.

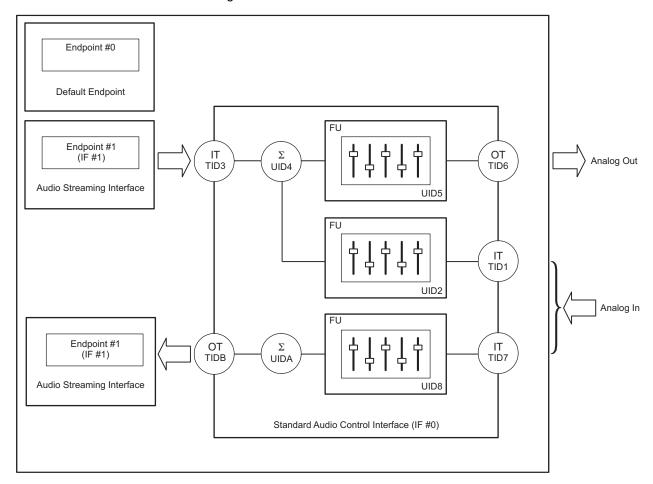


Figure 36. USB Audio Function Topology

#### 9.5.1.2 Interface #0

Interface #0 is the control interface. Alternative setting #0 is the only possible setting for interface #0. Alternative setting #0 describes the standard audio control interface. The audio control interface is constructed with a series of terminal connections. The PCM2912A has the following 10 terminals:

- Input terminal (Terminal ID#1) for audio analog input for sidetone
- Feature unit (Unit ID#2) for sidetone PGA
- Input terminal (Terminal ID#3) for isochronous out stream
- Mixer unit (Unit ID#4) for sidetone mixing
- Feature unit (Unit ID#5) for analog output PGA
- Output terminal (Terminal ID#6) for audio analog output
- Input terminal (Terminal ID#7) for audio analog input
- Feature unit (Unit ID#8) for analog input PGA
- Mixer unit (Unit ID#A) for analog input
- Output terminal (Terminal ID#B) for isochronous in stream



Input terminal #3 is defined as *USB* stream (terminal type 0x0101). Input terminal #3 can accept two-channel audio streams constructed by the left and right channels. Output terminal #6 is defined as a speaker (terminal type 0x0301). Input terminals #1 and #7 are defined as *Microphone* (terminal type 0x0201). Physically, these two input terminals are the same input, but logically duplicated. Output terminal B is defined as a *USB* stream (terminal type 0x0101). Output terminal B is a single-channel audio stream. Mixer unit #4 multiplexes the analog input (sidetone) and the audio data of the digital-to-analog converter (DAC). Mixer unit A is placed in front of output terminal B. Mixer unit A has no impact on recording data. Mixer units #4 and A do not have programming capability.

Feature unit #5 supports the following sound control features for analog outputs:

- Volume control
- Mute control

The built-in volume controller can be manipulated by an audio-class-specific request from 0 dB to -76 dB in steps of 1 dB. An individual (L and R) channel can be set for different values. The built-in mute controller can be manipulated by an audio-class-specific request. Only the master mute control request is acceptable.

Feature unit #2 supports the following sound control features for analog input (sidetone):

- Volume control
- Mute control

The built-in volume controller can be manipulated by an audio-class-specific request from 0 dB to -76 dB in 1-dB steps. Only the master volume control is acceptable. The built-in mute controller can be manipulated by audio-class-specific request. Only the master mute control request is acceptable.

Feature unit #8 supports the following sound control features for analog input (microphone record input):

- Volume control
- Mute control

The built-in analog volume controller can be manipulated by an audio-class-specific request from +30 dB to -12 dB in 1-dB steps. The built-in mute controller can be manipulated by an audio-class-specific request. Only the master mute control request is acceptable.

#### 9.5.1.3 Interface #1

Interface #1 is the audio streaming interface for data output. Table 4 lists the three alternative settings for Interface #1. Alternative setting #0 is the zero bandwidth setting. All other alternative settings are operational settings.

Table 4. Interface #1 Alternative Settings

ALTERNATIVE SETTING		DATA F	ORMAT	TRANSFER MODE	SAMPLING RATE (kHz)
00			Zero Band	lwidth	
01	16 bit Stereo 2s complement (PCM) Adaptive		Adaptive	8, 11.025, 16, 22.05, 32, 44.1, 48	
02	16 bit	Mono	2s complement (PCM)	Adaptive	8, 11.025, 16, 22.05, 32, 44.1, 48

## 9.5.1.4 Interface #2

Interface #2 is the audio streaming interface for data output. Table 5 shows the two alternative settings for Interface #2. Alternative setting #0 is the Zero Band Width setting. Alternative setting #1 is an operational setting.

**Table 5. Interface #2 Alternative Settings** 

ALTERNATIVE SETTING		DATA FO	PRMAT	TRANSFER MODE	SAMPLING RATE (kHz)
00			Zero Bandw	vidth	
01	16 bit	Mono	2s complement (PCM)	Asynchronous	8, 11.025, 16, 22.05, 32, 44.1, 48



#### 9.5.1.5 Endpoints

The PCM2912A has the following three endpoints:

- Control endpoint (EP #0)
- Isochronous out audio data stream endpoint (EP #1)
- Isochronous in audio data stream endpoint (EP #2)

The control endpoint is the default endpoint. The control endpoint controls all functions of the PCM2912A by the standard USB request and USB audio class-specific request from the host. The isochronous out audio data stream endpoint is an audio sink endpoint, which receives the PCM audio data. The isochronous out audio data stream endpoint accepts the asynchronous transfer mode. The isochronous in audio data stream endpoint is an audio source endpoint, which transmits the PCM audio data. The isochronous in audio data stream endpoint uses synchronous transfer mode.

#### 9.5.1.6 Internal Regulator

All required power sources are generated by five internal regulators.

Each regulator generates 3.3 V (typical, without load) from  $V_{BUS}$  (pin 2). Each regulator has an output pin and a ground return pin (as described in Table 6); this pair must be decoupled with an appropriate capacitor. Note that this capacitance affects inrush-current limitation. One band-gap reference circuit supplies reference voltage for all regulators. BGND (pin 1) is provided for reference ground of the band-gap reference.

**Table 6. Internal Regulator Summary** 

	_	•
SUPPLIED CIRCUIT	OUTPUT	RETURN
Digital	V <sub>DD</sub> (pin 5)	DGND (pin 6)
Analog	V <sub>CCA</sub> (pin 15)	AGND (pin 13)
Headphone (L-ch)	V <sub>CCL</sub> (pin 19)	HGND (pin 20)
Headphone (R-ch)	V <sub>CCR</sub> (pin 21)	HGND (pin 20)
PLL	V <sub>CCP</sub> (pin 26)	PGND (pin 25)



# 10 Application and Implementation

#### NOTE

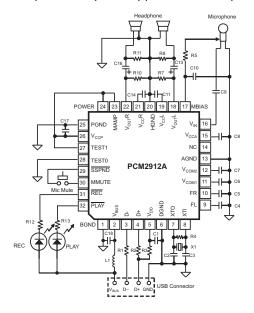
Information in the following applications sections is not part of the TI component specification, and TI does not warrant its accuracy or completeness. TI's customers are responsible for determining suitability of components for their purposes. Customers should validate and test their design implementation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

# 10.1 Application Information

The V<sub>BUS</sub> alerts the device when it has been plugged to a USB connection port. The /SSPND flag notifies when the USB input is idle for at least 5 ms; this flag can be used to control or notify subsequent circuits. More functional details can be found in *Interface Sequence*.

# 10.2 Typical Application

A bus-powered (Hi-power), +20-dB microphone amplifier application example is shown in Figure 37.



NOTE: X<sub>1</sub>: 6-MHz crystal resonator

 $C_1$ ,  $C_8$ ,  $C_{11}$ ,  $C_{14}$ ,  $C_{17}$ ,  $C_{18}$ : 1  $\mu F$  ceramic

C2, C3: 10 pF to 33 pF (depending on load capacitance of crystal resonator)

C<sub>4</sub>, C<sub>5</sub>: 100 pF ceramic

C<sub>6</sub>, C<sub>10</sub>: 3.3 µF

C<sub>7</sub>: 0.1 µF

 $C_9$ : 0.22  $\mu F$  electrolytic (depending on required frequency response for microphone input)

 $C_{13}$ ,  $C_{16}$ : 100  $\mu F$  electrolytic (depending on required frequency response for headphone output)

 $R_1$ ,  $R_2$ : 22  $\Omega$  to 33  $\Omega$ 

 $R_3$ : 1.5  $k\Omega$ 

R<sub>4</sub>: 1 MΩ

R<sub>5</sub>: 1 kΩ (depending on microphone characteristic)

 $R_7,\,R_8,\,R_{10},\,R_{11}$ : 3.3  $k\Omega$ 

 $R_{12}$ ,  $R_{13}$ : 820  $\Omega$  (depending on LED drive current)

 $L_1$ : 1  $\mu H$  (DC resistance < 0.6  $\Omega$ )

It is possible to change maximum power if total power of actual application does not require over 100 mA (set POWER = low to configure as low-power device).

Figure 37. USB Headset Application

Submit Documentation Feedback

Copyright © 2008–2015, Texas Instruments Incorporated



# **Typical Application (continued)**

#### **NOTE**

The circuit in Figure 37 is for information only. Total board design should be considered in order to meet the USB specification as a USB-compliant product.

#### 10.2.1 Design Requirements

For this design example, use the parameters listed in Table 7.

**Table 7. Design Parameters** 

DESIGN PARAMETER	EXAMPLE VALUE
Input voltage range	4.35 V to 5.25 V
Current	80 mA to 100 mA
Input clock frequency	11.994 MHz to 12.006 MHz

#### 10.2.2 Detailed Design Procedure

The PCM2912A is a simple design device, as can connect directly to a USB port. The device only requires decoupling capacitors on the voltage source pins and the output filter for the headphone amplifier; a recommended output filter is the one implemented in the PCM2912AEVM, shown in the user's guide (SBAU141).

# 10.2.3 Application Curves

For the application curves, see the graphs listed in Table 8.

**Table 8. Table of Graphs** 

		FIGURE
	Overall Characteristic	Figure 1
ADC Digital Decimation Filter Frequency	Stop Band Attenuation	Figure 2
Response	Passband Ripple	Figure 3
	Transient Band Response	Figure 4
ADC Digital High-Pass Filter Frequency	Stop Band Characteristic	Figure 5
Response	Passband Characteristic	Figure 6
ADC Analog Antialiasing Filter Frequency	Stop Band Characteristic	Figure 7
Response	Passband Characteristic	Figure 8
	Stop Band Attenuation	Figure 9
DAC Digital Interpolation Filter Frequency Response	Passband Ripple	Figure 10
response	Transient Band Response	Figure 11
DAC Analog FID Filter Fraguency Beanance	Stop Band Characteristic	Figure 12
DAC Analog FIR Filter Frequency Response	Passband Characteristic	Figure 13
DAC Analog Low-Pass Filter Frequency	Stop Band Characteristic	Figure 14
Response	Passband Characteristic	Figure 15
	THD+N at – 1 dB vs Temperature	Figure 16
	Dynamic Range and Signal-to-Noise Ratio vs Temperature	Figure 17
	THD+N at – 1 dB vs Supply Voltage	Figure 18
ADC	Dynamic Range and Signal-to-Noise Ratio vs Supply Voltage	Figure 19
	THD+N at – 1 dB vs Sampling Frequency	Figure 20
	Dynamic Range and Signal-to-Noise Ratio vs Sampling Frequency	Figure 21



# Table 8. Table of Graphs (continued)

		FIGURE
	THD+N at 0 dB vs Temperature	Figure 22
	Dynamic Range and Signal-to-Noise Ratio vs Temperature	Figure 23
	THD+N at 0 dB vs Supply Voltage	Figure 24
DAC	Dynamic Range and Signal-to-Noise Ratio vs Supply Voltage	Figure 25
	THD+N at 0 dB vs Sampling Frequency	Figure 26
	Dynamic Range and Signal-to-Noise Ratio vs Sampling Frequency	Figure 27
	Supply Current vs Supply Voltage	Figure 28
Supply Current	Supply Current vs Sampling Frequency	Figure 29
	Supply Current vs Temperature at Suspend Mode	Figure 30



# 11 Power Supply Recommendations

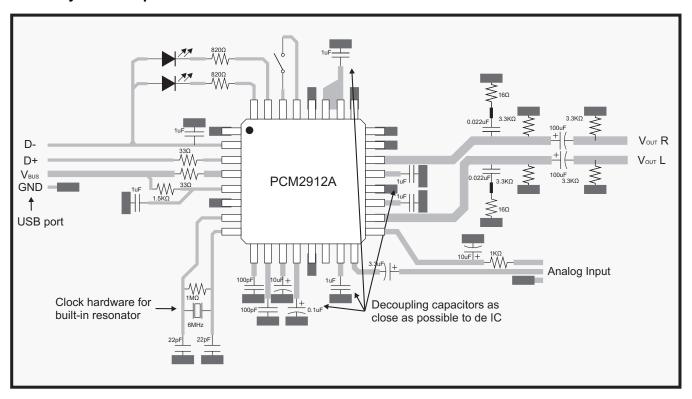
The voltage source to power the PCM2912A must be between 4.35 V and 5.25 V for proper operation (USB voltage). TI recommends placing a decoupling capacitor in every voltage source pin to help filter lower frequency power supply noise. Place these decoupling capacitors as close as possible to the PCM2912A.

# 12 Layout

# 12.1 Layout Guidelines

The decoupling capacitors must be as close as possible to the PCM2912A pins. TI recommends following the output filter for the headphone amplifier used in the PCM2912A EVM. The analog input needs a series capacitor to eliminate any possible offset level. The PCM2912A is a low-power device, thus there is no need for a special heat sink PCB design.

# 12.2 Layout Example



Connection to ground plane

Top layer traces

Top layer ground plane



# 13 Device and Documentation Support

#### 13.1 Documentation Support

#### 13.1.1 Related Documentation

For related documentation, see the following:

- Operating Envronments for PCM2912 Applications, SLAA387
- EVM User's Guide, SBAU141

#### 13.2 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.

**Design Support** *TI's Design Support* Quickly find helpful E2E forums along with design support tools and contact information for technical support.

#### 13.3 Trademarks

SpAct, E2E are trademarks of Texas Instruments. Audio Precision, System Two are trademarks of Audio Precision, Inc.. All other trademarks are the property of their respective owners.

#### 13.4 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.

## 13.5 Glossary

SLYZ022 — TI Glossarv.

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

# 14 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.



# PACKAGE OPTION ADDENDUM

10-Dec-2020

#### **PACKAGING INFORMATION**

www.ti.com

Orderable Device	Status (1)	Package Type	Package Drawing	Pins	Package Qty	Eco Plan	Lead finish/ Ball material	MSL Peak Temp	Op Temp (°C)	Device Marking (4/5)	Samples
PCM2912APJT	ACTIVE	TQFP	PJT	32	250	RoHS & Green	NIPDAU	Level-1-260C-UNLIM	-25 to 85	PCM2912A	Samples
PCM2912APJTR	ACTIVE	TQFP	PJT	32	1000	RoHS & Green	NIPDAU	Level-1-260C-UNLIM	-25 to 85	PCM2912A	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) RoHS: TI defines "RoHS" to mean semiconductor products that are compliant with the current EU RoHS requirements for all 10 RoHS substances, including the requirement that RoHS substance do not exceed 0.1% by weight in homogeneous materials. Where designed to be soldered at high temperatures, "RoHS" products are suitable for use in specified lead-free processes. TI may reference these types of products as "Pb-Free".

RoHS Exempt: TI defines "RoHS Exempt" to mean products that contain lead but are compliant with EU RoHS pursuant to a specific EU RoHS exemption.

**Green:** TI defines "Green" to mean the content of Chlorine (CI) and Bromine (Br) based flame retardants meet JS709B low halogen requirements of <=1000ppm threshold. Antimony trioxide based flame retardants must also meet the <=1000ppm threshold requirement.

- (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 finish/Ball material Orderable Devices may have multiple material finish options. Finish options are separated by a vertical ruled line. Lead finish/Ball material 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.





10-Dec-2020

# PACKAGE MATERIALS INFORMATION

www.ti.com 26-Feb-2019

# TAPE AND REEL INFORMATION





	Dimension designed to accommodate the component width
В0	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			Reel Diameter (mm)	Reel Width W1 (mm)	A0 (mm)	B0 (mm)	K0 (mm)	P1 (mm)	W (mm)	Pin1 Quadrant
PCM2912APJTR	TQFP	PJT	32	1000	330.0	16.4	9.6	9.6	1.5	12.0	16.0	Q2

**PACKAGE MATERIALS INFORMATION** 

www.ti.com 26-Feb-2019

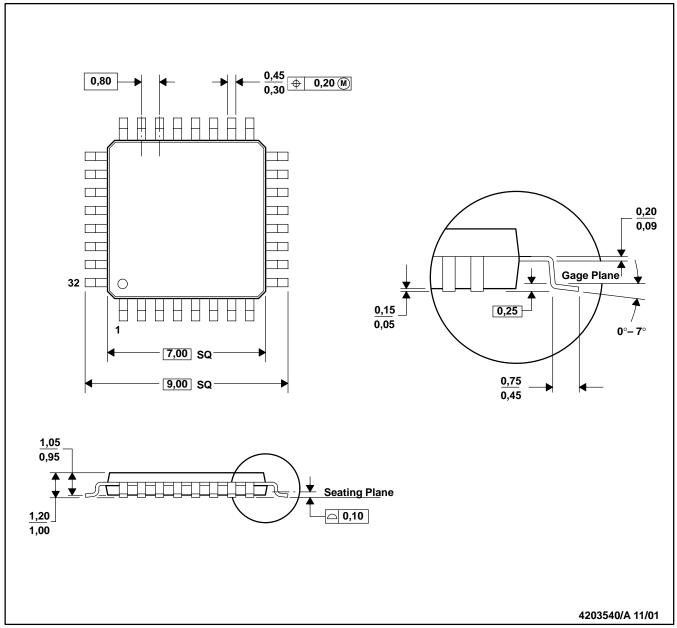


#### \*All dimensions are nominal

Device	Package Type	Package Drawing	Pins	SPQ	Length (mm)	Width (mm)	Height (mm)
PCM2912APJTR	TQFP	PJT	32	1000	350.0	350.0	43.0

# PJT (S-PQFP-N32)

#### PLASTIC QUAD FLATPACK



NOTES: A. All linear dimensions are in millimeters.

B. This drawing is subject to change without notice.

C. Falls within JEDEC MS-026

#### IMPORTANT NOTICE AND DISCLAIMER

TI PROVIDES TECHNICAL AND RELIABILITY DATA (INCLUDING DATASHEETS), DESIGN RESOURCES (INCLUDING REFERENCE DESIGNS), APPLICATION OR OTHER DESIGN ADVICE, WEB TOOLS, SAFETY INFORMATION, AND OTHER RESOURCES "AS IS" AND WITH ALL FAULTS, AND DISCLAIMS ALL WARRANTIES, EXPRESS AND IMPLIED, INCLUDING WITHOUT LIMITATION ANY IMPLIED WARRANTIES OF MERCHANTABILITY, FITNESS FOR A PARTICULAR PURPOSE OR NON-INFRINGEMENT OF THIRD PARTY INTELLECTUAL PROPERTY RIGHTS.

These resources are intended for skilled developers designing with TI products. You are solely responsible for (1) selecting the appropriate TI products for your application, (2) designing, validating and testing your application, and (3) ensuring your application meets applicable standards, and any other safety, security, or other requirements. These resources are subject to change without notice. TI grants you permission to use these resources only for development of an application that uses the TI products described in the resource. Other reproduction and display of these resources is prohibited. No license is granted to any other TI intellectual property right or to any third party intellectual property right. TI disclaims responsibility for, and you will fully indemnify TI and its representatives against, any claims, damages, costs, losses, and liabilities arising out of your use of these resources.

Tl's products are provided subject to Tl's Terms of Sale (<a href="www.ti.com/legal/termsofsale.html">www.ti.com/legal/termsofsale.html</a>) or other applicable terms available either on ti.com or provided in conjunction with such Tl products. Tl's provision of these resources does not expand or otherwise alter Tl's applicable warranties or warranty disclaimers for Tl products.

Mailing Address: Texas Instruments, Post Office Box 655303, Dallas, Texas 75265 Copyright © 2020, Texas Instruments Incorporated