

SLES265A - MARCH 2011 - REVISED AUGUST 2012

# SINGLE-ENDED, ANALOG-INPUT 24-BIT, 96-kHz STEREO A/D CONVERTER

Check for Samples: PCM1808-Q1

# **FEATURES**

- **Qualified for Automotive Applications** ٠
- 24-Bit Delta-Sigma Stereo A/D Converter
- Single-Ended Voltage Input: 3 Vp-p
- **High Performance:** 
  - THD + N: –93 dB (Typical)
  - SNR: 99 dB (Typical)
  - Dynamic Range: 99 dB (Typical)
- **Oversampling Decimation Filter:** 
  - Oversampling Frequency: ×64
  - Pass-Band Ripple: ±0.05 dB
  - Stop-Band Attenuation: –65 dB
  - On-Chip High-Pass Filter: 0.91 Hz (48 kHz)
- Flexible PCM Audio Interface
  - Master/Slave Mode Selectable
  - Data Formats: 24-Bit I<sup>2</sup>S, 24-Bit Left-Justified
- Power Down and Reset by Halting System Clock
- Analog Antialias LPF Included
- Sampling Rate: 8 kHz-96 kHz
- System Clock: 256  $f_S$ , 384  $f_S$ , 512  $f_S$
- **Dual Power Supplies:** 
  - 5-V for Analog
  - 3.3-V for Digital

### Package: 14-Pin TSSOP

## DESCRIPTION

The PCM1808-Q1 is high-performance, low-cost, single-chip, stereo analog-to-digital converter with single-ended analog voltage input. The PCM1808-Q1 uses a delta-sigma modulator with 64-times oversampling and includes a digital decimation filter and high-pass filter that removes the dc component of the input signal. For various applications, the PCM1808-Q1 supports master and slave mode and two data formats in serial audio interface.

The PCM1808-Q1 supports the power-down and reset function by means of halting the system clock.

The PCM1808-Q1 is suitable for wide variety of costsensitive consumer applications where aood performance and operation with a 5-V analog supply and 3.3-V digital supply is required. The PCM1808-Q1 is fabricated using a highly advanced CMOS process and is available in a small, 14-pin TSSOP package.



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# PCM1808-Q1



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This integrated circuit can be damaged by ESD. Texas Instruments recommends that all integrated circuits be handled with appropriate precautions. Failure to observe proper handling and installation procedures can cause damage.

ESD damage can range from subtle performance degradation to complete device failure. Precision integrated circuits may be more susceptible to damage because very small parametric changes could cause the device not to meet its published specifications.

## **ABSOLUTE MAXIMUM RATINGS**

over operating free-air temperature range (unless otherwise noted) <sup>(1)</sup>

	PCM1808-Q1
Analog supply voltage, V <sub>CC</sub>	–0.3 V to 6.5 V
Digital supply voltage, V <sub>DD</sub>	–0.3 V to 4 V
Ground voltage differences, AGND, DGND	±0.1 V
Digital input voltage, LRCK, BCK, DOUT	-0.3 V to (V <sub>DD</sub> + 0.3 V) < 4 V
Digital input voltage, SCKI, MD0, MD1, FMT	–0.3 V to 6.5 V
Analog input voltage, V <sub>IN</sub> L, V <sub>IN</sub> R, V <sub>REF</sub>	-0.3 V to (V <sub>CC</sub> + 0.3 V) < 6.5 V
Input current (any pins except supplies)	±10 mA
Ambient temperature under bias, T <sub>A</sub>	–40°C to 125°C
Storage temperature, T <sub>stg</sub>	–55°C to 150°C
Junction temperature, T <sub>J</sub>	150°C
Lead temperature (soldering)	260°C, 5 s
Package temperature (reflow, peak)	260°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.

# **RECOMMENDED OPERATING CONDITIONS**

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

		MIN	NOM	MAX	UNIT
Analog supply voltage, V <sub>CC</sub>		4.5	5	5.5	V
Digital supply voltage, V <sub>DD</sub>		2.7	3.3	3.6	V
Analog input voltage, full scale (-0 dB)	$V_{CC} = 5 V$	2.93	3	3.23	Vp-p
Digital input logic family		TTL	TTL compatible		
Digital input clock frequency, system clock		2.048		49.152	MHz
Digital input clock frequency, sampling clock		8		96	kHz
Digital output load capacitance				20	pF
Operating free-air temperature, T <sub>A</sub>		-40		125	°C



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### **ELECTRICAL CHARACTERISTICS**

All specifications at  $T_A = 25^{\circ}C$ ,  $V_{CC} = 5 \text{ V}$ ,  $V_{DD} = 3.3 \text{ V}$ , master mode,  $f_S = 48 \text{ kHz}$ , system clock = 512  $f_S$ , 24-bit data, unless otherwise noted

	PARAMETER	TEST CON	DITIONS	MIN	TYP	MAX	UNIT
	Resolution				24		Bits
DATA FO	RMAT	T					
	Audio data interface format			I <sup>2</sup> S,	left-justified		
	Audio data bit length				24		Bits
	Audio data format			MSB-firs	t, 2s comple	ement	
f <sub>S</sub>	Sampling frequency			8	48	96	kHz
		256 f <sub>S</sub>		2.048	12.288	24.576	
	System clock frequency, -40°C $\leq T_A \leq 125°C^{(1)}$	384 f <sub>S</sub>		3.072	18.432	36.864	MHz
	40 0 = 1A = 120 0	512 f <sub>S</sub>		4.096	24.576	49.152	
NPUT LC	OGIC						
V <sub>IH</sub> <sup>(2)</sup>				2		$V_{DD}$	
V <sub>IL</sub> <sup>(2)</sup>	put logic level,			0		0.8	
V <sub>IH</sub> <sup>(4)</sup> <sup>(5)</sup>	$-40^{\circ}C \le T_{A} \le 125^{\circ}C^{(3)}$			2		5.5	VDC
V <sub>IL</sub> <sup>(4)</sup> <sup>(5)</sup>	)			0		0.8	
ін <sup>(4)</sup>		$V_{IN} = V_{DD}$				±10	
IL <sup>(4)</sup>	Input logic current	V <sub>IN</sub> = 0 V				±10	
		., .,	at 25°C		65	100	μA
(2) (5) IH		$V_{IN} = V_{DD}$	–40°C ≤ T <sub>A</sub> ≤ 125°C		65	150	
IL <sup>(2) (5)</sup>	-	V <sub>IN</sub> = 0 V				±10	
OUTPUT	LOGIC						
. (6)			at 25°C	2.8			VDC
V <sub>ОН</sub> <sup>(6)</sup>	Output logic level <sup>(3)</sup>	$I_{OUT} = -4 \text{ mA}$	–40°C ≤ T <sub>A</sub> ≤ 125°C	2.7			
V <sub>OL</sub> <sup>(6)</sup>		$I_{OUT} = 4 \text{ mA}, -40^{\circ}\text{C} \le \text{T}_{\text{A}} \le$				0.5	
	JRACY, –40°C ≤ T <sub>A</sub> ≤ 125°C	<u> </u>					
	Gain mismatch, channel-to- channel				±1	±3	% of FSR
	Gain error				±3	±6	% of FSR
DYNAMIC	PERFORMANCE <sup>(7)</sup>					·	
			at 25°C		-93	-87	-
		$V_{IN}$ = -0.5 dB, f <sub>S</sub> = 48 kHz	$-40^{\circ}C \le T_A \le 125^{\circ}C$		-93	-85	
THD + N	Total harmonic distortion + noise	$V_{IN} = -0.5 \text{ dB}, \text{ f}_{S} = 96 \text{ kHz}$	(8)		-87		dB
		$V_{IN} = -60 \text{ dB}, \text{ f}_{S} = 48 \text{ kHz}$			-37		
		$V_{IN} = -60 \text{ dB}, f_S = 96 \text{ kHz}$	(8)		-39		
		at 25°C		95	99		
	Dynamic range	$f_{S}$ = 48 kHz, A-weighted	$-40^{\circ}C \le T_A \le 125^{\circ}C$	93	99		dB
	-	$f_{S}$ = 96 kHz, A-weighted <sup>(8)</sup>			101		
			at 25°C	95	99		
S/N	Signal-to-noise ratio	$f_{S}$ = 48 kHz, A-weighted	$-40^{\circ}C \le T_A \le 125^{\circ}C$	93	99		dB
S/N Signal-to-hoise ratio		f <sub>S</sub> = 96 kHz, A-weighted <sup>(8)</sup>			101		

(1)  $384 f_s$  where  $f_s = 96 kHz$ , and 512 where  $f_s = 48 kHz$  and 96 kHz are functionally tested. Other options are specified by design.

Pins 7, 8: LRCK, BCK (Schmitt-trigger input, with 50-kΩ typical pulldown resistor, in slave mode) (2)

(3) Specified by design

(4) Pin 6: SCKI (Schmitt-trigger input, 5-V tolerant)
 (5) Pins 10–12: MD0, MD1, FMT (Schmitt-trigger input, with 50-kΩ typical pulldown resistor, 5-V tolerant)

(6) Pins 7-9: LRCK, BCK (in master mode), DOUT

Analog performance specifications are tested using a System Two™ audio measurement system by Audio Precision™ with 400-Hz HPF (7) and 20-kHz LPF in RMS mode.

 $f_S = 96$  kHz, system clock = 256  $f_S$ . (8)

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# **ELECTRICAL CHARACTERISTICS (continued)**

All specifications at  $T_A = 25^{\circ}$ C,  $V_{CC} = 5$  V,  $V_{DD} = 3.3$  V, master mode,  $f_S = 48$  kHz, system clock = 512  $f_S$ , 24-bit data, unless otherwise noted

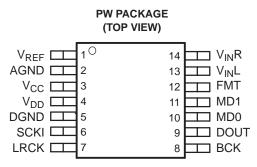
	PARAMETER	TEST	CONDITIONS	MIN	ТҮР	MAX	UNIT
		f 40 kl la	at 25°C	93	97		
	Channel separation	f <sub>S</sub> = 48 kHz	–40°C ≤ T <sub>A</sub> ≤ 125°C	91	97		dB
		f <sub>S</sub> = 96 kHz <sup>(8)</sup>	L		91		
ANALC	DG INPUT						
	Input voltage, –40°C ≤ T <sub>A</sub> ≤ 125°C			0.58 V <sub>CC</sub>	0.6 V <sub>CC</sub>	0.65 V <sub>CC</sub>	Vp-р
	Center voltage input range, $-40^{\circ}C \le T_A \le 125^{\circ}C$			0.2 V <sub>CC</sub>	0.5 V <sub>CC</sub>	0.8 V <sub>CC</sub>	V
	Input impedance				60		kΩ
	Antialiasing filter frequency response	–3 dB			1.3		MHz
DIGITA	L FILTER PERFORMANCE, -4	0°C ≤ T <sub>A</sub> ≤ 125°C <sup>(3)</sup>					
	Pass band					0.454 f <sub>S</sub>	Hz
	Stop band			0.583 f <sub>S</sub>			Hz
	Pass-band ripple					±0.05	dB
	Stop-band attenuation			-65			dB
	Delay time				17.4/f <sub>S</sub>		
	HPF frequency response	–3 dB			0.019 f <sub>S</sub> /1000		
POWE	R SUPPLY REQUIREMENTS			-		÷	
V <sub>CC</sub>	Voltage range,			4.5	5	5.5	VDC
V <sub>DD</sub>	–40°C ≤ T <sub>A</sub> ≤ 125°C			2.7	3.3	3.6	VDC
		f <sub>S</sub> = 48 kHz, 96 kHz, –	40°C ≤ T <sub>A</sub> ≤ 125°C <sup>(10)</sup>		8.6	11	mA
CC		Powered down (11)			1		μA
	Supply current <sup>(9)</sup>	6 40 111-	at 25°C		5.9	8	
	Supply current (*)	f <sub>S</sub> = 48 kHz	–40°C ≤ T <sub>A</sub> ≤ 125°C		5.9	10	mA
DD		$f_{S} = 96 \text{ kHz}^{(10)}$			10.2		mA
		Powered down (11)			150		μA
		f <sub>S</sub> = 48 kHz			62	81	m\//
	Power dissipation <sup>(9)</sup>	$f_{\rm S} = 96 \text{ kHz}^{(10)}$			77		mW
		Powered down (11)			500		μW
ТЕМРЕ	ERATURE RANGE						
T <sub>A</sub>	Operation temperature			-40		125	°C
θ <sub>JA</sub>	Thermal resistance				170		°C/W

(9) Minimum load on LRCK (pin 7), BCK (pin 8), DOUT (pin 9) (10)  $f_S = 96$  kHz, system clock = 256  $f_S$ . (11) Power-down and reset functions enabled by halting SCKI, BCK, LRCK.



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## **PIN ASSIGNMENTS**



P0032-02

#### **TERMINAL FUNCTIONS**

TERM	IINAL	I/O	DESCRIPTION
NAME	PIN		
AGND	2	_	Analog GND
BCK	8	I/O	Audio data bit clock input/output <sup>(1)</sup>
DGND	5	_	Digital GND
DOUT	9	0	Audio data digital output
FMT	12	I	Audio interface format select <sup>(2)</sup>
LRCK	7	I/O	Audio data latch enable input/output <sup>(1)</sup>
MD0	10	I	Audio interface mode select 0 <sup>(2)</sup>
MD1	11	I	Audio interface mode select 1 <sup>(2)</sup>
SCKI	6	I	System clock input; 256 $f_S$ , 384 $f_S$ or 512 $f_S$ <sup>(3)</sup>
V <sub>CC</sub>	3	_	Analog power supply, 5-V
V <sub>DD</sub>	4	_	Digital power supply, 3.3-V
V <sub>IN</sub> L	13	I	Analog input, L-channel
V <sub>IN</sub> R	14	I	Analog input, R-channel
V <sub>REF</sub>	1	-	Reference voltage decoupling (= 0.5 V <sub>CC</sub> )

(1)

Schmitt-trigger input with internal pulldown (50-k $\Omega$ , typical) Schmitt-trigger input with internal pulldown (50-k $\Omega$ , typical), 5-V tolerant Schmitt-trigger input, 5-V tolerant

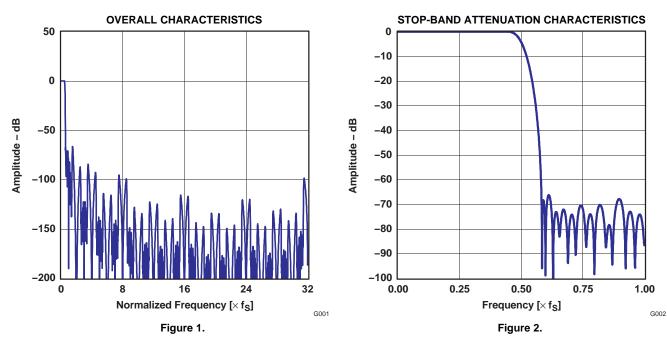
(2) (3)

Antialias Delta-Sigma BCK VINL LPF Modulator LRCK Serial ×1/64 Interface DOUT Decimation V<sub>REF</sub> Reference Filter FMT with Mode/ **High-Pass Filter** Format Control MD1 Delta-Sigma Antialias MD0 VINR LPF Modulator **Clock and Timing Control** SCKI Power Supply V<sub>CC</sub> AGND DGND V<sub>DD</sub> B0004-10

## **TYPICAL PERFORMANCE CURVES OF INTERNAL FILTERS**

All specifications at  $T_A = 25^{\circ}$ C,  $V_{CC} = 5$  V,  $V_{DD} = 3.3$  V, master mode,  $f_S = 48$  kHz, system clock = 512  $f_S$ , 24-bit data, unless otherwise noted.

# DECIMATION FILTER FREQUENCY RESPONSE



Functional Block Diagram

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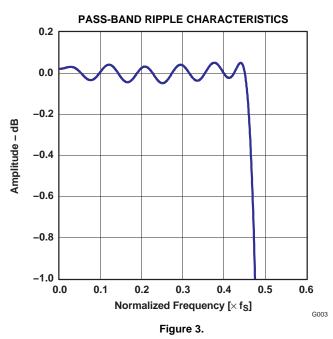


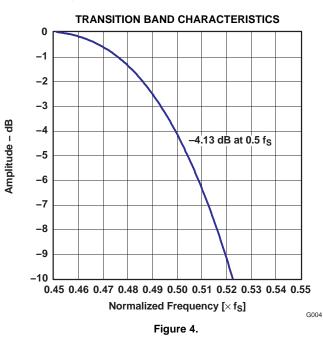
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## **TYPICAL PERFORMANCE CURVES OF INTERNAL FILTERS (Continued)**

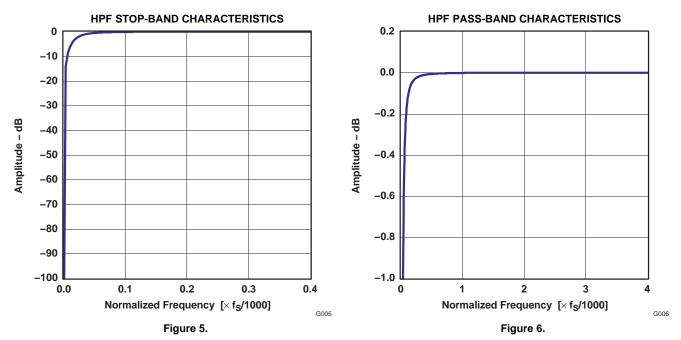
All specifications at  $T_A = 25^{\circ}C$ ,  $V_{CC} = 5$  V,  $V_{DD} = 3.3$  V, master mode,  $f_S = 48$  kHz, system clock = 512  $f_S$ , 24-bit data, unless otherwise noted.

### **DECIMATION FILTER FREQUENCY RESPONSE (Continued)**





### HIGH-PASS FILTER FREQUENCY RESPONSE

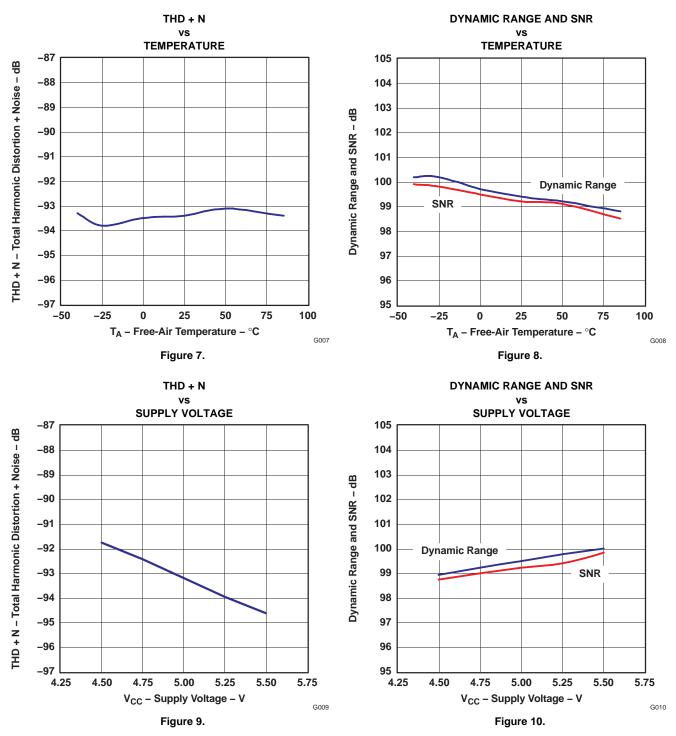


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### **TYPICAL PERFORMANCE CURVES**

All specifications at  $T_A = 25^{\circ}C$ ,  $V_{CC} = 5 \text{ V}$ ,  $V_{DD} = 3.3 \text{ V}$ , master mode,  $f_S = 48 \text{ kHz}$ , system clock = 512  $f_S$ , 24-bit data, unless otherwise noted.

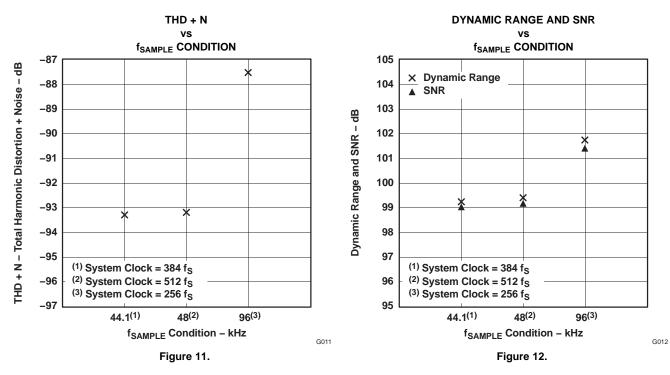




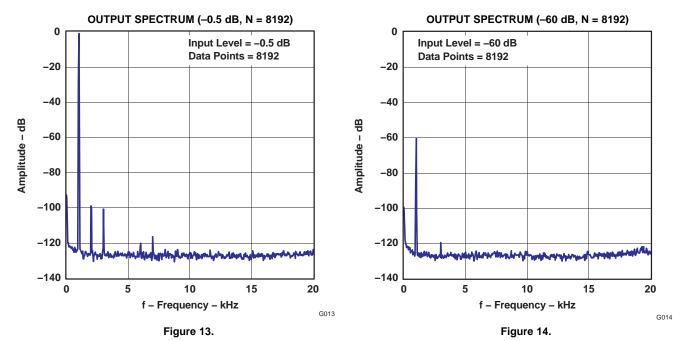
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#### **TYPICAL PERFORMANCE CURVES (Continued)**

All specifications at  $T_A = 25^{\circ}C$ ,  $V_{CC} = 5 \text{ V}$ ,  $V_{DD} = 3.3 \text{ V}$ , master mode,  $f_S = 48 \text{ kHz}$ , system clock = 512  $f_S$ , 24-bit data, unless otherwise noted.



#### **OUTPUT SPECTRUM**



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# PCM1808-Q1

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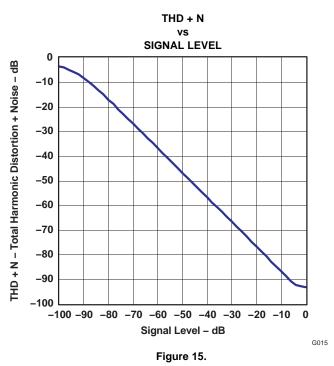


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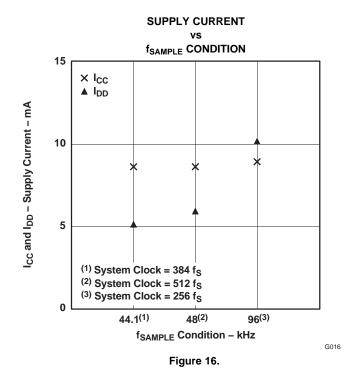
# **TYPICAL PERFORMANCE CURVES (Continued)**

All specifications at  $T_A = 25^{\circ}$ C,  $V_{CC} = 5$  V,  $V_{DD} = 3.3$  V, master mode,  $f_S = 48$  kHz, system clock = 512  $f_S$ , 24-bit data, unless otherwise noted.

## **OUTPUT SPECTRUM (Continued)**



### SUPPLY CURRENT



### SYSTEM CLOCK

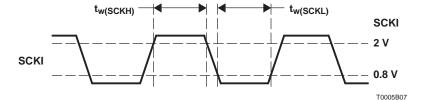
The PCM1808-Q1 supports 256  $f_S$ , 384  $f_S$  and 512  $f_S$  as system clock, where  $f_S$  is the audio sampling frequency. The system clock must be supplied on SCKI (pin 6).

The PCM1808-Q1 has a system clock detection circuit which automatically senses if the system clock is operating at 256  $f_S$ , 384  $f_S$ , or 512  $f_S$  in slave mode. In master mode, the system clock frequency must be controlled through the serial control port, which uses MD1 (pin 111) and MD0 (pin 10). The system clock is divided down automatically to generate frequencies of 128  $f_S$  and 64  $f_S$ , which are used to operate the digital filter and the delta-sigma modulator, respectively.

Table 1 shows some typical relationships between sampling frequency and system clock frequency, and Figure 17 shows system clock timing.

SAMPLING FREQUENCY (kHz)	SYSTEM	I CLOCK FREQUENCY (f <sub>SC</sub>	<sub>LK</sub> ) (MHz)
	256 f <sub>S</sub>	384 f <sub>S</sub>	512 f <sub>S</sub>
8	2.048	3.072	4.096
16	4.096	6.144	8.192
32	8.192	12.288	16.384
44.1	11.2896	16.9344	22.5792
48	12.288	18.432	24.576
64	16.384	24.576	32.768
88.2	22.5792	33.8688	45.1584
96	24.576	36.864	49.152

Table 1. Sampling Frequency and System Clock Frequency



SYMBOL	PARAMETER	MIN	MAX	UNIT
t <sub>w(SCKH)</sub>	System clock pulse duration, HIGH	8		ns
t <sub>w(SCKL)</sub>	System clock pulse duration, LOW	8		ns
	System clock duty cycle	40%	60%	

#### Figure 17. System Clock Timing

### FADE-IN AND FADE-OUT FUNCTIONS

The PCM1808-Q1 has fade-in and fade-out functions on DOUT (pin 9) to avoid pop noise, and the functions come into operation in some cases as described in several following sections. The level changes from 0 dB to mute or mute to 0 dB are performed using calculated pseudo S-shaped characteristics with zero-cross detection. Because of the zero-cross detection, the time needed for the fade in and fade out depends on the analog input frequency ( $f_{in}$ ). It takes 48/ $f_{in}$  until processing is completed. If there is no zero cross during 8192/ $f_{S}$ , DOUT is faded in or out by force during 48/ $f_{S}$  (TIME OUT). Figure 18 illustrates the fade-in and fade-out operation processing.

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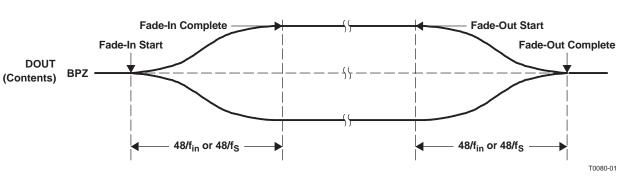
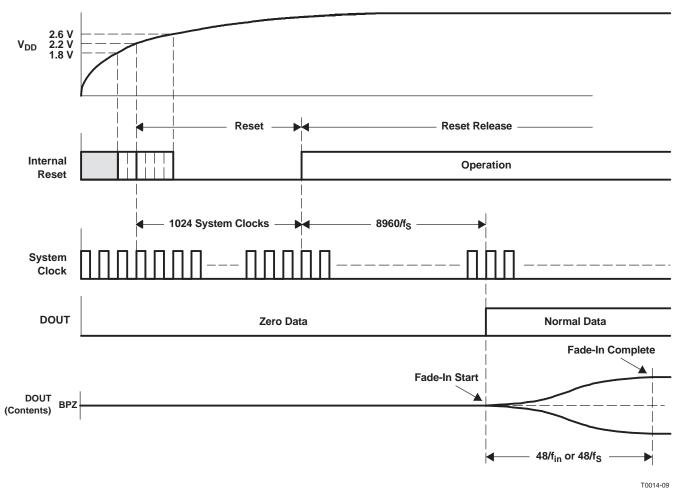
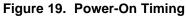


Figure 18. Fade-In and Fade-Out Operations

## **POWER ON**

The PCM1808-Q1 has an internal power-on-reset circuit, and initialization (reset) is performed automatically when the power supply ( $V_{DD}$ ) exceeds 2.2 V (typical). While  $V_{DD} < 2.2$  V (typical), and for 1024 system-clock counts after  $V_{DD} > 2.2$  V (typical), the PCM1808-Q1 stays in the reset state and the digital output is forced to zero. The digital output is valid after the reset state is released and the time of 8960/f<sub>S</sub> has elapsed. Because the fade-in operation is performed, it takes additional time of 48/f<sub>in</sub> or 48/f<sub>S</sub> until the data corresponding to the analog input signal is obtained. Figure 19 illustrates the power-on timing and the digital output.







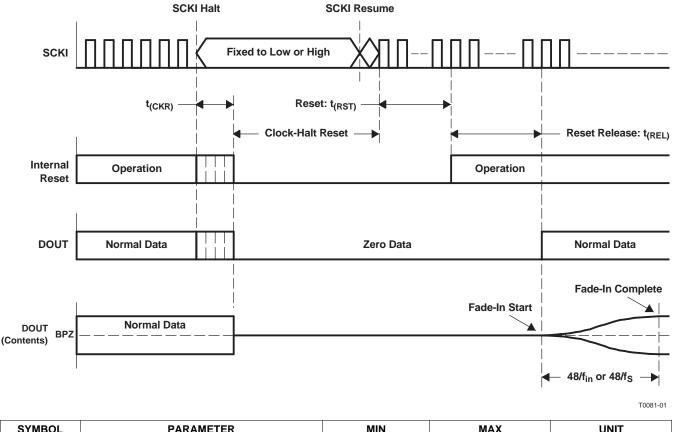
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#### **CLOCK-HALT POWER-DOWN AND RESET FUNCTION**

The PCM1808-Q1 has a power-down and reset function, which is triggered by halting SCKI (pin 6) in both master and slave modes. The function is available anytime after power on. Reset and power down are performed automatically 4  $\mu$ s (minimum) after SCKI is halted. While the clock-halt reset is asserted, the PCM1808-Q1 stays in the reset and power-down mode, and DOUT (pin 9) is forced to zero. SCKI must be supplied to release the reset and power-down mode. The digital output is valid after the reset state is released and the time of 1024 SCKI + 8960/f<sub>S</sub> has elapsed. Because the fade-in operation is performed, it takes additional time of 48/f<sub>in</sub> or 48/f<sub>S</sub> until the level corresponding to the analog input signal is obtained. Figure 20 illustrates the clock-halt reset timing.

To avoid ADC performance degradation, BCK (pin 8) and LRCK (pin 7) are required to synchronize with SCKI within  $4480/f_S$  after SCKI is resumed. If it takes more than  $4480/f_S$  for BCK and LRCK to synchronize with SCKI, SCKI should be masked until the synchronization is achieved again, taking care of glitch and jitter. See the typical circuit connection diagram, Figure 26.

To avoid ADC performance degradation, the clock-halt reset also should be asserted when system clock SCKIor the audio interface clocks BCK and LRCK (sampling rate  $f_S$ ) are changed on the fly.



SYMBOL	PARAMETER	MIN	MAX	UNIT
t <sub>(CKR)</sub>	Delay time from SCKI halt to internal reset	4		μs
t <sub>(RST)</sub>	Delay time from SCKI resume to reset release		1024 SCKI	μs
t <sub>(REL)</sub>	Delay time from reset release to DOUT output		8960/f <sub>S</sub>	μs

Figure 20. Clock-Halt Power-Down and Reset Timing



## SERIAL AUDIO DATA INTERFACE

The PCM1808-Q1 interfaces the audio system through LRCK (pin 7), BCK (pin 8), and DOUT (pin 9).

#### INTERFACE MODE

The PCM1808-Q1 supports master mode and slave mode as interface modes, which are selected by MD1 (pin 11) and MD0 (pin 10), as shown in Table 2. MD1 and MD0 must be set prior to power on.

In master mode, the PCM1808-Q1 provides the timing of serial audio data communications between the PCM1808-Q1 and the digital audio processor or external circuit. While in slave mode, the PCM1808-Q1 receives the timing for data transfer from an external controller.

MD1 (Pin 11)	MD0 (Pin 10)	INTERFACE MODE	
Low	Low	Slave mode (256 $f_S$ , 384 $f_S$ , 512 $f_S$ autodetection)	
Low	High	Master mode (512 f <sub>S</sub> )	
High	Low	Master mode (384 f <sub>S</sub> )	
High	High	Master mode (256 f <sub>S</sub> )	

 Table 2. Interface Modes

#### Master mode

In master mode, BCK and LRCK work as output pins, and these pins are controlled by timing which is generated in the clock circuit of the PCM1808-Q1. The frequency of BCK is fixed at 64 BCK/frame.

#### Slave mode

In slave mode, BCK and LRCK work as input pins. The PCM1808-Q1 accepts 64-BCK/frame or 48-BCK/frame format (only for a 384-f<sub>s</sub> system clock), not 32-BCK/frame format.

#### DATA FORMAT

The PCM1808-Q1 supports two audio data formats in both master and slave modes. The data formats are selected by FMT (pin 12), as shown in Table 3. Figure 21 illustrates the data formats in slave mode and master mode.

#### Table 3. Data Format

FORMAT NO.	FMT (Pin 12)	FORMAT
0	Low	l <sup>2</sup> S, 24-bit
1	High	Left-justified, 24-bit



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#### FORMAT 0: FMT = LOW

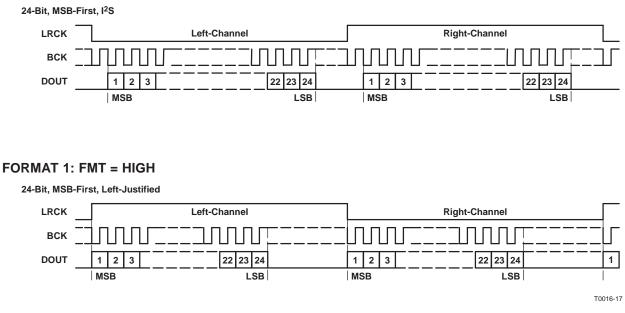
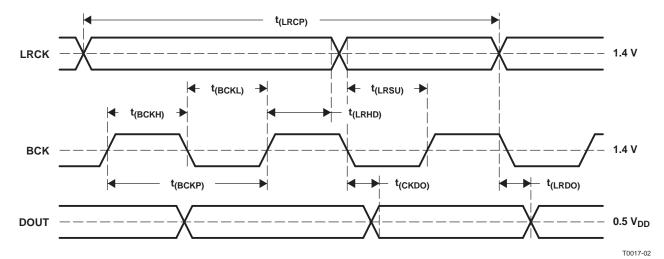


Figure 21. Audio Data Format (LRCK and BCK Work as Inputs in Slave Mode and as Outputs in Master Mode)



### **INTERFACE TIMING**

Figure 22 and Figure 23 illustrate the interface timing in slave mode and master mode, respectively.



SYMBOL	PARAMETER	MIN	TYP	MAX	UNIT
t <sub>(BCKP)</sub>	BCK period	1/(64 f <sub>S</sub> )			ns
t <sub>(BCKH)</sub>	BCK pulse duration, HIGH	1.5 × t <sub>(SCKI)</sub>			ns
t <sub>(BCKL)</sub>	BCK pulse duration, LOW	1.5 × t <sub>(SCKI)</sub>			ns
t <sub>(LRSU)</sub>	LRCK setup time to BCK rising edge	50			ns
t <sub>(LRHD)</sub>	LRCK hold time to BCK rising edge	10			ns
t <sub>(LRCP)</sub>	LRCK period	10			μs
t <sub>(CKDO)</sub>	Delay time, BCK falling edge to DOUT valid	-10		40	ns
t <sub>(LRDO)</sub>	Delay time, LRCK edge to DOUT valid	-10		40	ns
t <sub>r</sub>	Rise time of all signals			20	ns
t <sub>f</sub>	Fall time of all signals			20	ns

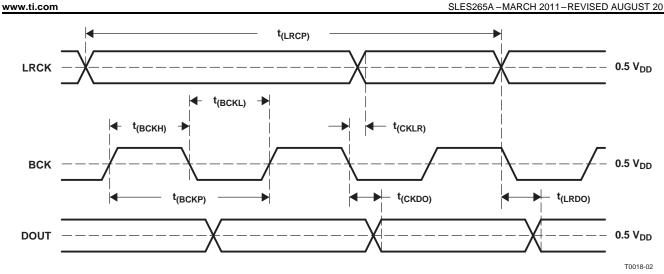
NOTE: Timing measurement reference level is 1.4 V for input and 0.5 V<sub>DD</sub> for output. Rise and fall times are from 10% to 90% of the input/output signal swing. Load capacitance of DOUT is 20 pF. t<sub>(SCKI)</sub> is the SCKI period.

#### Figure 22. Audio Data Interface Timing (Slave Mode: LRCK and BCK Work as Inputs)



# PCM1808-Q1

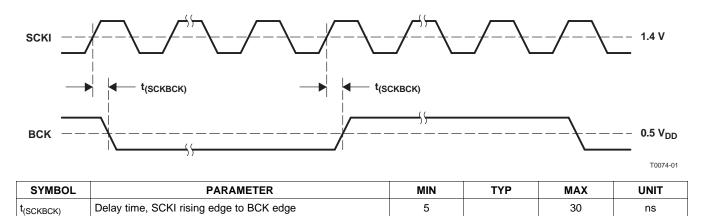
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SYMBOL	PARAMETER	MIN	TYP	MAX	UNIT
t <sub>(BCKP)</sub>	BCK period	150	1/(64 f <sub>S</sub> )	2000	ns
t <sub>(BCKH)</sub>	BCK pulse duration, HIGH	65		1200	ns
t <sub>(BCKL)</sub>	BCK pulse duration, LOW	65		1200	ns
t <sub>(CKLR)</sub>	Delay time, BCK falling edge to LRCK valid	-10		20	ns
t <sub>(LRCP)</sub>	LRCK period	10	1/f <sub>S</sub>	125	μs
t <sub>(CKDO)</sub>	Delay time, BCK falling edge to DOUT valid	-10		20	ns
t <sub>(LRDO)</sub>	Delay time, LRCK edge to DOUT valid	-10		20	ns
t <sub>r</sub>	Rise time of all signals			20	ns
t <sub>f</sub>	Fall time of all signals			20	ns

NOTE: Timing measurement reference level is 0.5 V<sub>DD</sub>. Rise and fall times are from 10% to 90% of the input/output signal swing. Load capacitance of all signals is 20 pF.

#### Figure 23. Audio Data Interface Timing (Master Mode: LRCK and BCK Work as Outputs)



NOTE: Timing measurement reference level is 1.4 V for input and 0.5 V<sub>DD</sub> for output. Load capacitance of BCK is 20 pF. This timing is applied when SCKI frequency is less than 25 MHz.

#### Figure 24. Audio Clock Interface Timing (Master Mode: BCK Works as Output)

# SYNCHRONIZATION WITH DIGITAL AUDIO SYSTEM

In slave mode, the PCM1808-Q1 operates under LRCK (pin 7), synchronized with system clock SCKI (pin 6). The PCM1808-Q1 does not require a specific phase relationship between LRCK and SCKI, but does require the synchronization of LRCK and SCKI.

If the relationship between LRCK and SCKI changes more than  $\pm 6$  BCKs for 64 BCK/frame ( $\pm 5$  BCKs for 48 BCK/frame) during one sample period due to LRCK or SCKI jitter, internal operation of the ADC halts within  $1/f_s$  and digital output is forced to zero data (BPZ code) until resynchronization between LRCK and SCKI is established.

In the case of changes less than ±5 BCKs for 64 BCK/frame (±4 BCKs for 48 BCK/frame), resynchronization does not occur and the previously described digital output control and discontinuity do not occur.

Figure 25 illustrates the digital output response for loss of synchronization and resynchronization. During undefined data, the PCM1808-Q1 can generate some noise in the audio signal. Also, the transition of normal data to undefined data creates a discontinuity in the digital output data, which can generate some noise in the audio signal. The digital output is valid after resynchronization completes and the time of  $32/f_S$  has elapsed. Because the fade-in operation is performed, it takes additional time of  $48/f_{in}$  or  $48/f_S$  until the level corresponding to the analog input signal is obtained. If synchronization is lost during the fade-in operation, the operation stops and DOUT (pin 9) is forced to zero data immediately. The fade-in operation resumes from mute after the time of  $32/f_S$  following resynchronization.

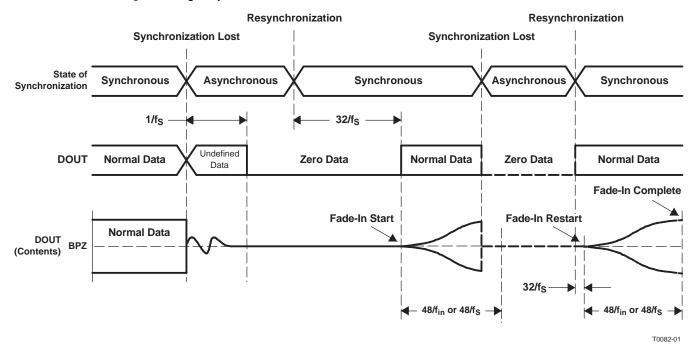


Figure 25. ADC Digital Output for Loss of Synchronization and Resynchronization

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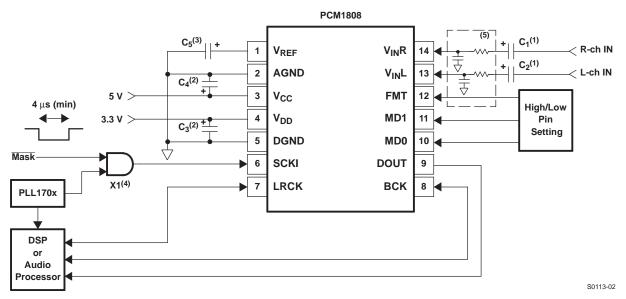


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### **APPLICATION INFORMATION**

#### **TYPICAL CIRCUIT CONNECTION DIAGRAM**

Figure 26 is a typical circuit connection diagram. The antialiasing low-pass filters are integrated on the analog inputs,  $V_{IN}L$  and  $V_{IN}R$ . If the performance of these filters is not adequate for an application, appropriate external antialiasing filters are needed. A passive RC filter (100  $\Omega$  and 0.01  $\mu$ F to 1 k $\Omega$  and 1000 pF) generally is used.



- (1) C1, C2: A 1- $\mu$ F electrolytic capacitor gives 2.7 Hz ( $\tau = 1 \mu$ F × 60 k $\Omega$ ) cutoff frequency for the input HPF in normal operation and requires a power-on settling time with a 60-ms time constant in the power-on initialization period.
- (2) C3, C4: Bypass capacitors, 0.1-µF ceramic and 10-µF electrolytic, depending on layout and power supply
- (3) C5:  $0.1-\mu F$  ceramic and  $10-\mu F$  electrolytic capacitors are recommended.
- (4) X1: X1 masks the system clock input when using the clock-halt reset function with external control.
- (5) Optional external antialiasing filter could be required, depending on the application.

#### Figure 26. Typical Circuit Connection Diagram

### BOARD DESIGN AND LAYOUT CONSIDERATIONS

#### V<sub>CC</sub>, V<sub>DD</sub> PINS

The digital and analog power supply lines to the PCM1808-Q1 should be bypassed to the corresponding ground pins with both 0.1- $\mu$ F ceramic and 10- $\mu$ F electrolytic capacitors as close to the pins as possible to maximize the dynamic performance of the ADC.

#### AGND, DGND PINS

To maximize the dynamic performance of the PCM1808-Q1, the analog and digital grounds are not internally connected. These grounds should have low impedance to avoid digital noise feedback into the analog ground. They should be connected directly to each other under the PCM1808-Q1 package to reduce potential noise problems.

### V<sub>IN</sub>L, V<sub>IN</sub>R PINS

 $V_{IN}L$  and  $V_{IN}R$  are single-ended inputs. The antialias low-pass filters are integrated on these inputs to remove the high-frequency noise outside the audio band. If the performance of these filters is not adequate for an application, appropriate external antialiasing filters are required. A passive RC filter (100  $\Omega$  and 0.01  $\mu$ F to 1 k $\Omega$  and 1000 pF) is generally used.



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## $V_{\text{REF}}$ PIN

To ensure low source impedance of the ADC references, 0.1- $\mu$ F ceramic and 10- $\mu$ F electrolytic capacitors are recommended between V<sub>REF</sub> and AGND. These capacitors should be located as close as possible to the V<sub>REF</sub> pin to reduce dynamic errors on the ADC references.

### DOUT PIN

The DOUT pin has a large load-drive capability, but if the DOUT line is long, locating a buffer near the PCM1808-Q1 and minimizing load capacitance is recommended to minimize the digital-analog crosstalk and maximize the dynamic performance of the ADC.

### SYSTEM CLOCK

The quality of the system clock can influence dynamic performance, as the PCM1808-Q1 operates based on a system clock. Therefore, it may be necessary to consider the system clock duty, jitter, and the time difference between system clock transition and BCK or LRCK transition in slave mode.

# PCM1808-Q1

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# **REVISION HISTORY**

### Changes from Original (March, 2011) to Revision A

•	ROC CHANGES: Added 2.93 min and 3.23 max to analog input voltage row	2
•	ELEC CHAR CHANGES: Added -40°C $\leq$ T <sub>A</sub> $\leq$ 125°C to the header for DC accuracy and the rows for system clock frequency, input logic level, and output logic level	. 3
•	Added test condition row to $V_{IN} = V_{DD}$ (input logic current) at -40°C $\leq T_A \leq 125$ °C with typ value 65 and max value 150	. 3
•	Added test condition row to $I_{OUT} = -4$ mA (output logic level) at $-40^{\circ}C \le T_A \le 125^{\circ}C$ with a min value of 2.7; added test condition of 25°C with min value of 2.8	. 3
•	Added test condition of 25°C to $V_{IN} = -0.5 \text{ dB}$ , $f_S = 48 \text{ kHz}$ (THD + N) with max value of -87, and added row with test condition of -40°C $\leq T_A \leq 125$ °C and max value -85	. 3
•	Added test condition 25°C and min value of 95; added test condition row for $-40^{\circ}C \le T_A \le 125^{\circ}C$ with min value of 93 to $f_S = 48$ kHz, A-weighted row (dynamic range and signal-to-noise)	. 3
•	Added test condition 25°C and min value of 93; added test condition row for $-40^{\circ}C \le T_{A} \le 125^{\circ}C$ with min value of 91 to $f_{S} = 48$ kHz (channel separation)	. 4
•	Added min value 0.58 $V_{CC}$ and max value 0.65 $V_{CC}$ to input voltage; added 0.2 $V_{CC}$ min and 0.8 $V_{CC}$ max to center voltage; changed center voltage Vref to center voltage input range	. 4
•	Added -40°C $\leq$ T <sub>A</sub> $\leq$ 125°C to input voltage, center voltage, digital filter performance header, supply current, and voltage range rows	. 4
•	Added test condition row with -40°C $\leq T_A \leq 125$ °C to $f_S = 48$ kHz (supply current) with a typ value of 5.9 and a max value of 10	. 4



6-Feb-2020

# PACKAGING INFORMATION

Orderable Device	Status	Package Type	Package	Pins	Package	Eco Plan	Lead/Ball Finish	MSL Peak Temp	Op Temp (°C)	Device Marking	Samples
	(1)		Drawing		Qty	(2)	(6)	(3)		(4/5)	
PCM1808QPWRQ1	ACTIVE	TSSOP	PW	14	2000	Green (RoHS & no Sb/Br)	NIPDAU	Level-3-260C-168 HR	-40 to 125	P1808Q	Samples

<sup>(1)</sup> 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.

<sup>(2)</sup> 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.

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

<sup>(4)</sup> 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.

<sup>(6)</sup> 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.

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#### OTHER QUALIFIED VERSIONS OF PCM1808-Q1 :



# PACKAGE OPTION ADDENDUM

6-Feb-2020

• Catalog: PCM1808

NOTE: Qualified Version Definitions:

Catalog - TI's standard catalog product

# PACKAGE MATERIALS INFORMATION

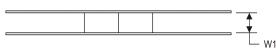
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## TAPE AND REEL INFORMATION

#### REEL DIMENSIONS

Texas Instruments





#### 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

## TAPE AND REEL INFORMATION

\*All dimensions are nominal

Device	Package Type	Package Drawing		SPQ	Reel Diameter (mm)	Reel Width W1 (mm)	· · /	B0 (mm)	K0 (mm)	P1 (mm)	W (mm)	Pin1 Quadrant
PCM1808QPWRQ1	TSSOP	PW	14	2000	330.0	12.4	6.9	5.6	1.6	8.0	12.0	Q1

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# PACKAGE MATERIALS INFORMATION

31-Jul-2012



\*All dimensions are nominal

Device	Package Type	Package Drawing	Pins	SPQ	Length (mm)	Width (mm)	Height (mm)
PCM1808QPWRQ1	TSSOP	PW	14	2000	367.0	367.0	35.0

PW (R-PDSO-G14)

PLASTIC SMALL OUTLINE



A. An integration of the information o

Body length does not include mold flash, protrusions, or gate burrs. Mold flash, protrusions, or gate burrs shall not exceed 0,15 each side.

Body width does not include interlead flash. Interlead flash shall not exceed 0,25 each side.

E. Falls within JEDEC MO-153





NOTES: A. All linear dimensions are in millimeters.

- B. This drawing is subject to change without notice.
- C. Publication IPC-7351 is recommended for alternate designs.
- D. Laser cutting apertures with trapezoidal walls and also rounding corners will offer better paste release. Customers should contact their board assembly site for stencil design recommendations. Refer to IPC-7525 for other stencil recommendations.
- E. Customers should contact their board fabrication site for solder mask tolerances between and around signal pads.



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