Burr-Brown Products
from Texas Instruments

PCM1770, PCM1771

SLES011E−SEPTEMBER 2001−REVISED MARCH 2007

LOW-VOLTAGE AND LOW-POWER STEREO AUDIO DIGITAL-TO-ANALOG CONVERTER WITH HEADPHONE AMPLIFIER

FEATURES

- **Multilevel DAC Including Headphone Amplifier**
- **Analog Performance (V_{CC}, V_{HP} = 2.4 V):**
  - **Dynamic Range: 98 dB Typ**
  - **THD+N at 0 dB: 0.1% Typ**
  - **THD+N at −20 dB: 0.04% Typ**
  - **Output Power at R_L = 16 Ω: 13 mW (Stereo), 26 mW (Monaural)**
- **1.6-V to 3.6-V Single Power Supply**
- **Low Power Dissipation: 6.5 mW at V_{CC}, V_{HP} = 2.4 V**
- **System Clock: 128 f_S, 192 f_S, 256 f_S, 384 f_S**
- **Sampling Frequency: 5 kHz to 50 kHz**
- **Software Control (PCM1770):**
  - **16-, 20-, 24-Bit Word Available**
  - **Left-, Right-Justified, and I²S**
  - **Slave/Master Selectable**
  - **Digital Attenuation: 0 dB to −62 dB, 1 dB/Step**
  - **44.1-kHz Digital De-Emphasis**
  - **Zero Cross Attenuation**
  - **Digital Soft Mute**
  - **Monaural Analog-In With Mixing**
  - **Monaural Speaker Mode**
- **Hardware Control (PCM1771):**
  - **Left-Justified and I²S**
  - **44.1-kHz Digital De-Emphasis**
  - **Monaural Analog-In With Mixing**
- **Pop-Noise-Free Circuit**
- **3.3-V Tolerant**
- **Packages: TSSOP-16 and VQFN-20**

APPLICATIONS

- **Portable Audio Player**
- **Cellular Phone**
- **PDA**
- **Other Applications Requiring Low-Voltage Operation**

DESCRIPTION

The PCM1770 and PCM1771 devices are CMOS, monolithic, integrated circuits which include stereo digital-to-analog converters, headphone circuitry, and support circuitry in small TSSOP-16 and VQFN-20 packages.

The data converters use TI's enhanced multilevel Δ-Σ architecture, which employs noise shaping and multilevel amplitude quantization to achieve excellent dynamic performance and improved tolerance to clock jitter. The PCM1770 and PCM1771 devices accept several industry standard audio data formats with 16- to 24-bit data, left-justified, I²S, etc., providing easy interfacing to audio DSP and decoder devices. Sampling rates up to 50 kHz are supported. A full set of user-programmable functions is accessible through a 3-wire serial control port, which supports register write functions.

Please be aware that an important notice concerning availability, standard warranty, and use in critical applications of Texas Instruments semiconductor products and disclaimers thereto appears at the end of this data sheet.

PRODUCTION DATA information is current as of publication date. Products conform to specifications per the terms of the Texas Instruments standard warranty. Production processing does not necessarily include testing of all parameters.

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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.

ABSOLUTE MAXIMUM RATINGS

over operating free-air temperature range unless otherwise noted[1]

	PCM1770 PCM1771
Supply voltage: V_{CC}, V_{HP}	−0.3 V to 4 V
Supply voltage differences: V_{CC}, V_{HP}	±0.1 V
Ground voltage differences	±0.1 V
Digital input voltage	−0.3 V to 4 V
Input current (any terminals except supplies)	±10 mA
Operating temperature	−40°C to 125°C
Storage temperature	−55°C to 150°C
Junction temperature	150°C
Lead temperature (soldering)	260°C, 5 s
Package temperature (IR 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

		MIN	NOM	MAX	UNIT
Supply voltage: V_{CC}, V_{HP}		1.6	2.4	3.6	V
Digital input logic family			CMOS		
Digital input clock frequency	System clock	0.64		19.2	MHz
	Sampling clock	5		50	kHz
Analog output load resistance		16			Ω
Analog input level (V_{HP} = 2.4 V)				1.4	Vp-p
Operating free-air temperature, T_A		−25		85	°C

ELECTRICAL CHARACTERISTICS

all specifications at T_A = 25°C, V_{CC} = V_{HP} = 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S and 24-bit data, R_L = 16 Ω, unless otherwise noted

PARAMETER		TEST CONDITIONS	PCM1770PW, PCM1771PW, PCM1770RGA, PCM1771RGA			UNIT
			MIN	TYP	MAX	
	Resolution			24		Bits
OPERATING FREQUENCY						
	Sampling frequency (f_S)		5		50	kHz
	System clock frequency		128f_S, 192 f_S, 256 f_S, 384 f_S			
DIGITAL INPUT/OUTPUT[1][2]						
	Logic family	CMOS compatible				
V_{IH}	Input logic level		0.7 V_{CC}			VDC
V_{IL}					0.3 V_{CC}	VDC
I_{IH}	Input logic current	V_{IN} = V_{CC}			10	µA
I_{IL}		V_{IN} = 0 V			−10	µA
V_{OH}	Output logic level[3]	I_{OH} = −2 mA	0.7 V_{CC}			VDC
V_{OL}		I_{OL} = 2 mA			0.3 V_{CC}	VDC
DYNAMIC PERFORMANCE (HEADPHONE OUTPUT)						
	Full-scale output voltage	0 dB		0.55 V_{HP}		Vp-p
	Dynamic range	EIAJ, A-weighted	90	98		dB
	Signal-to-noise ratio	EIAJ, A-weighted	90	98		dB
THD+N	Total harmonic distortion + noise	0 dB (13 mW)		0.1%		
		−20 dB (0.1 mW)		0.04%	0.1%	
	Output power	Stereo	10	13		mWrms
		Monaural	20	26		mWrms
	Channel separation		64	72		dB
	Load resistance		14	16		Ω
DC ACCURACY						
	Gain error			±2	±8	%FSR
	Gain mismatch, channel-to-channel			±2	±8	%FSR
	Bipolar zero error	V_{OUT} = 0.5 V_{CC} at BPZ		±30	±75	mV
ANALOG LINE INPUT (MIXING CIRCUIT)						
	Analog input voltage range				0.584 V_{HP}	Vp-p
	Gain (analog input to headphone output)			0.67		
	Analog input impedance			10		kΩ
THD+N	Total harmonic distortion + noise	AIN = 0.56 V_{HP} (peak-to-peak)		0.1%		
DIGITAL FILTER PERFORMANCE						
	Pass band				0.454 f_S	
	Stop band		0.546 f_S			
	Pass-band ripple				±0.04	dB
	Stop-band attenuation		−50			dB
	Group delay			20/f_S		
	44.1-kHz de-emphasis error			±0.1		dB
ANALOG FILTER PERFORMANCE						

(1) Digital inputs and outputs are CMOS compatible.
(2) All logic inputs are 3.3-V tolerant and not terminated internally.
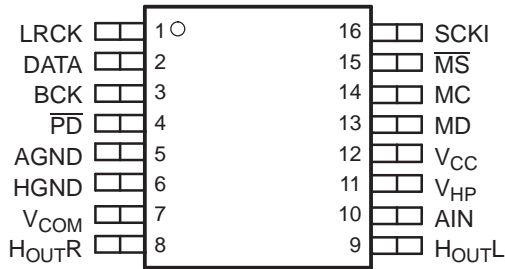(3) LRCK and BCK terminals

ELECTRICAL CHARACTERISTICS (continued)

all specifications at T_A = 25°C, V_{CC} = V_{HP} = 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S and 24-bit data, R_L = 16 Ω, unless otherwise noted

PARAMETER		TEST CONDITIONS	PCM1770PW, PCM1771PW, PCM1770RGA, PCM1771RGA			UNIT
			MIN	TYP	MAX	
Frequency response		at 20 kHz		±0.2		dB
POWER SUPPLY REQUIREMENTS						
Voltage range, V_{CC}, V_{HP}			1.6	2.4	3.6	VDC
I_{CC}	Supply current	BPZ input		1.5	2.5	mA
I_{HP}		BPZ input		1.2	2.5	
$I_{CC} + I_{HP}$		Power down[4]		5	15	µA
	Power dissipation	BPZ input		6.5	12	mW
		Power down[4]		12	36	µW
TEMPERATURE RANGE						
Operation temperature			−25		85	°C
$θ_{JA}$	Thermal resistance	PCM1770PW, -71PW: 16-terminal TSSOP		150		°C/W
		PCM1770RGA, -71RGA: 20-terminal VQFN		130		

(4) All input signals are held static.

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PIN ASSIGNMENTS

PCM1770
PW PACKAGE
(TOP VIEW)

LRCK	1 ○	16	SCKI
DATA	2	15	$\overline{MS}$
BCK	3	14	MC
$\overline{PD}$	4	13	MD
AGND	5	12	V_{CC}
HGND	6	11	V_{HP}
V_{COM}	7	10	AIN
$H_{OUT}R$	8	9	$H_{OUT}L$

PCM1771
PW PACKAGE
(TOP VIEW)

LRCK	1 ○	16	SCKI
DATA	2	15	FMT
BCK	3	14	AMIX
$\overline{PD}$	4	13	DEMP
AGND	5	12	V_{CC}
HGND	6	11	V_{HP}
V_{COM}	7	10	AIN
$H_{OUT}R$	8	9	$H_{OUT}L$
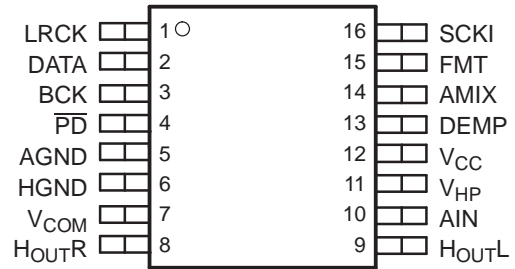
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PCM1770
RGA PACKAGE
(TOP VIEW)

Top pins: LRCK (20), NC (19), NC (18), NC (17), SCKI (16)

DATA	1	15	$\overline{MS}$
BCK	2	14	MC
$\overline{PD}$	3	13	MD
AGND	4	12	V_{CC}
HGND	5	11	V_{HP}

Bottom pins: V_{COM} (6), $H_{OUT}R$ (7), NC (8), $H_{OUT}L$ (9), AIN (10)

PCM1771
RGA PACKAGE
(TOP VIEW)

Top pins: LRCK (20), NC (19), NC (18), NC (17), SCKI (16)

DATA	1	15	FMT
BCK	2	14	AMIX
$\overline{PD}$	3	13	DEMP
AGND	4	12	V_{CC}
HGND	5	11	V_{HP}

Bottom pins: V_{COM} (6), $H_{OUT}R$ (7), NC (8), $H_{OUT}L$ (9), AIN (10)
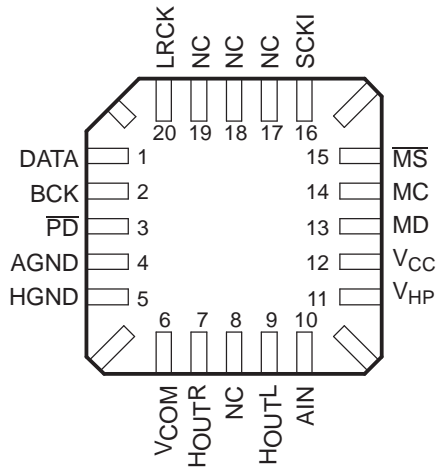
NC − No internal connection

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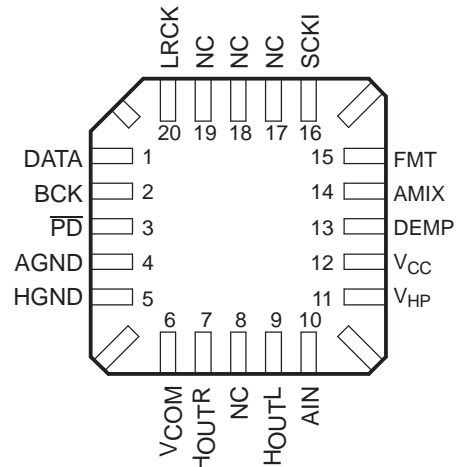
TERMINAL FUNCTIONS

PCM1770PW

| TERMINAL NAME	NO.	I/O	DESCRIPTION
AGND	5	−	Analog ground. This is a return for V_{CC}.
AIN	10	I	Monaural analog signal mixer input. The signal can be mixed with the outputs of the L- and R-channel DACs.
BCK	3	I/O	Serial bit clock. Clocks the individual bits of the audio data input, DATA. In the slave interface mode, this clock is input from an external device. In the master interface mode, the PCM1770 device generates the BCK output to an external device.
DATA	2	I	Serial audio data input
HGND	6	−	Analog ground. This is a return for V_{HP}.
$H_{OUT}L$	9	O	L-channel analog signal output of the headphone amplifiers
$H_{OUT}R$	8	O	R-channel analog signal output of the headphone amplifiers
LRCK	1	I/O	Left and right clock. Determines which channel is being input on the audio data input, DATA. The frequency of LRCK must be the same as the audio sampling rate. In the slave interface mode, this clock is input from an external device. In the master interface mode, the PCM1770 device generates the LRCK output to an external device.
MC	14	I	Mode control port serial bit clock input. Clocks the individual bits of the control data input, MD.
MD	13	I	Mode control port serial data input. Controls the operation mode on the PCM1770 device.
$\overline{MS}$	15	I	Mode control port select. The control port is active when this terminal is low.
$\overline{PD}$	4	I	Reset input. When low, the PCM1770 device is powered down, and all mode control registers are reset to default settings.
SCKI	16	I	System clock input
V_{CC}	12	−	Power supply for all analog circuits except the headphone amplifier.
V_{COM}	7	−	Decoupling capacitor connection. An external 10-μF capacitor connected from this terminal to analog ground is required for noise filtering. Voltage level of this terminal is 0.5 V_{HP} nominal.
V_{HP}	11	−	Analog power supply for the headphone amplifier circuits. The voltage level must be the same as V_{CC}.

TERMINAL FUNCTIONS (CONTINUED)

PCM1770RGA

TERMINAL NAME	NO.	I/O	DESCRIPTION
AGND	4	−	Analog ground. This is a return for V_{CC}.
AIN	10	I	Monaural analog signal mixer input. The signal can be mixed with the outputs of the L- and R-channel DACs.
BCK	2	I/O	Serial bit clock. Clocks the individual bits of the audio data input, DATA. In the slave interface mode, this clock is input from an external device. In the master interface mode, the PCM1770 device generates the BCK output to an external device.
DATA	1	I	Serial audio data input
HGND	5	−	Analog ground. This is a return for V_{HP}.
$H_{OUT}L$	9	O	L-channel analog signal output of the headphone amplifiers
$H_{OUT}R$	7	O	R-channel analog signal output of the headphone amplifiers
LRCK	20	I/O	Left and right clock. Determines which channel is being input on the audio data input, DATA. The frequency of LRCK must be the same as the audio sampling rate. In the slave interface mode, this clock is input from an external device. In the master interface mode, the PCM1770 device generates the LRCK output to an external device.
MC	14	I	Mode control port serial bit clock input. Clocks the individual bits of the control data input, MD.
MD	13	I	Mode control port serial data input. Controls the operation mode on the PCM1770 device.
$\overline{MS}$	15	I	Mode control port select. The control port is active when this terminal is low.
NC	8, 17, 18, 19	−	No connect
$\overline{PD}$	3	I	Reset input. When low, the PCM1770 device is powered down, and all mode control registers are reset to default settings.
SCKI	16	I	System clock input
V_{CC}	12	−	Power supply for all analog circuits except the headphone amplifier.
V_{COM}	6	−	Decoupling capacitor connection. An external 10-μF capacitor connected from this terminal to analog ground is required for noise filtering. Voltage level of this terminal is 0.5 V_{HP} nominal.
V_{HP}	11	−	Analog power supply for the headphone amplifier circuits. The voltage level must be the same as V_{CC}.

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TERMINAL FUNCTIONS (CONTINUED)

PCM1771PW

TERMINAL NAME	NO.	I/O	DESCRIPTION
AGND	5	–	Analog ground. This is a return for V_{CC}.
AIN	10	–	Monaural analog signal mixer input. The signal can be mixed with the outputs of the L- and R-channel DACs.
AMIX	14	I	Analog mixing control
BCK	3	I	Serial bit clock. Clocks the individual bits of the audio data input, DATA.
DATA	2	I	Serial audio data input
DEMP	13	I	De-emphasis control
FMT	15	I	Data format select
HGND	6	–	Analog ground. This is a return for V_{HP}.
$H_{OUT}L$	9	O	L-channel analog signal output of the headphone amplifiers
$H_{OUT}R$	8	O	R-channel analog signal output of the headphone amplifiers
LRCK	1	I	Left and right clock. Determines which channel is being input on the audio data input, DATA. The frequency of LRCK must be the same as the audio sampling rate.
$\overline{PD}$	4	I	Reset input. When low, the PCM1771 device is powered down, and all mode control registers are reset to default settings.
SCKI	16	I	System clock input
V_{CC}	12	–	Power supply for all analog circuits except the headphone amplifier.
V_{COM}	7	–	Decoupling capacitor connection. An external 10-µF capacitor connected from this terminal to analog ground is required for noise filtering. Voltage level of this terminal is 0.5 V_{HP} nominal.
V_{HP}	11	–	Analog power supply for the headphone amplifier circuits. The voltage level must be the same as V_{CC}.

PCM1771RGA

TERMINAL NAME	NO.	I/O	DESCRIPTION
AGND	4	–	Analog ground. This is a return for V_{CC}.
AIN	10	–	Monaural analog signal mixer input. The signal can be mixed with the outputs of the L- and R-channel DACs.
AMIX	14	I	Analog mixing control
BCK	2	I	Serial bit clock. Clocks the individual bits of the audio data input, DATA.
DATA	1	I	Serial audio data input
DEMP	13	I	De-emphasis control
FMT	15	I	Data format select
HGND	5	–	Analog ground. This is a return for V_{HP}.
$H_{OUT}L$	9	O	L-channel analog signal output of the headphone amplifiers
$H_{OUT}R$	7	O	R-channel analog signal output of the headphone amplifiers
LRCK	20	I	Left and right clock. Determines which channel is being input on the audio data input, DATA. The frequency of LRCK must be the same as the audio sampling rate.
NC	8, 17, 18, 19	–	No connect
$\overline{PD}$	3	I	Reset input. When low, the PCM1771 device is powered down, and all mode control registers are reset to default settings.
SCKI	16	I	System clock input
V_{CC}	12	–	Power supply for all analog circuits except the headphone amplifier
V_{COM}	6	–	Decoupling capacitor connection. An external 10-µF capacitor connected from this terminal to analog ground is required for noise filtering. Voltage level of this terminal is 0.5 V_{HP} nominal.
V_{HP}	11	–	Analog power supply for the headphone amplifier circuits. The voltage level must be the same as V_{CC}.

FUNCTIONAL BLOCK DIAGRAM
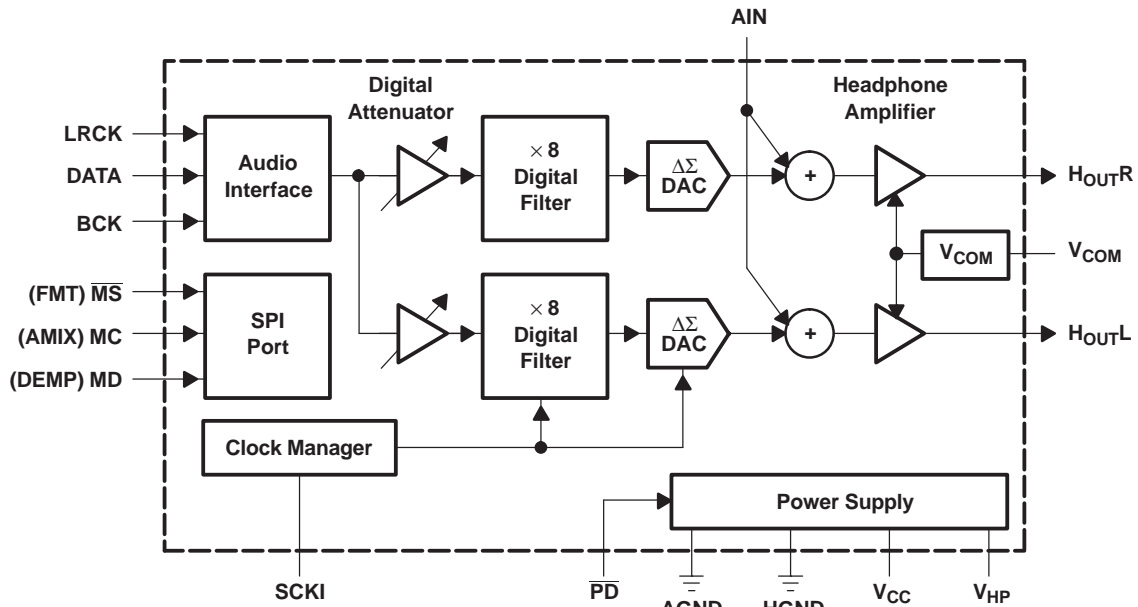
( ) : PCM1771

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TYPICAL PERFORMANCE CURVES

All specifications at T_A = 25°C, V_{CC} = V_{HP} 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S, 24-bit data, and R_L = 16 Ω, unless otherwise noted.

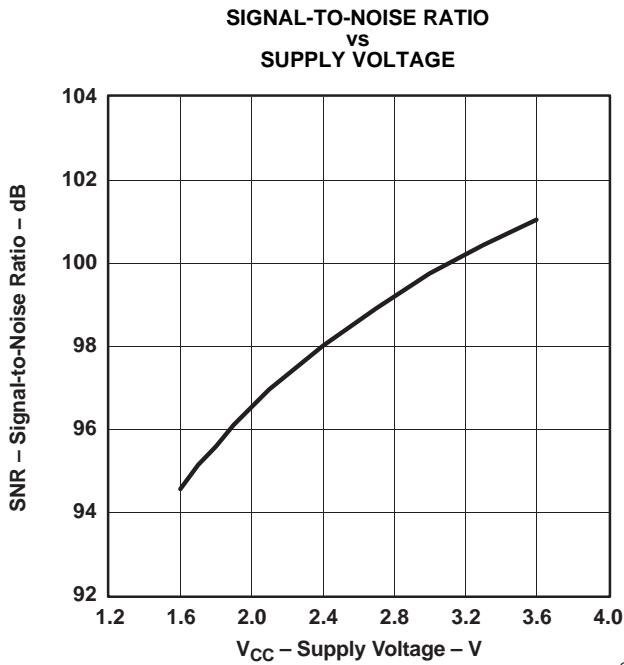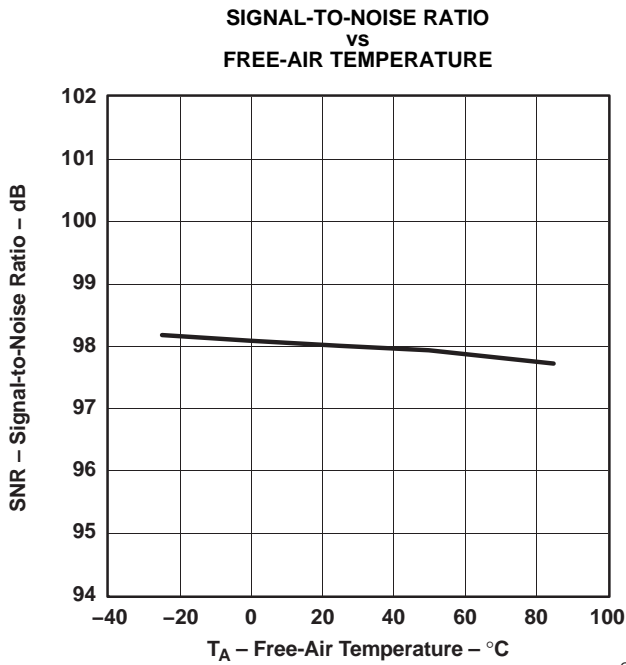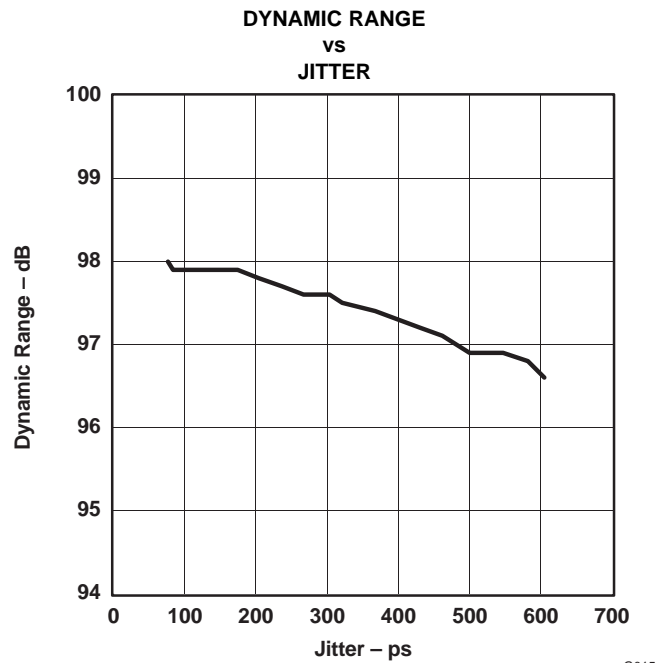## Digital Filter

Digital Filter (De-Emphasis Off)
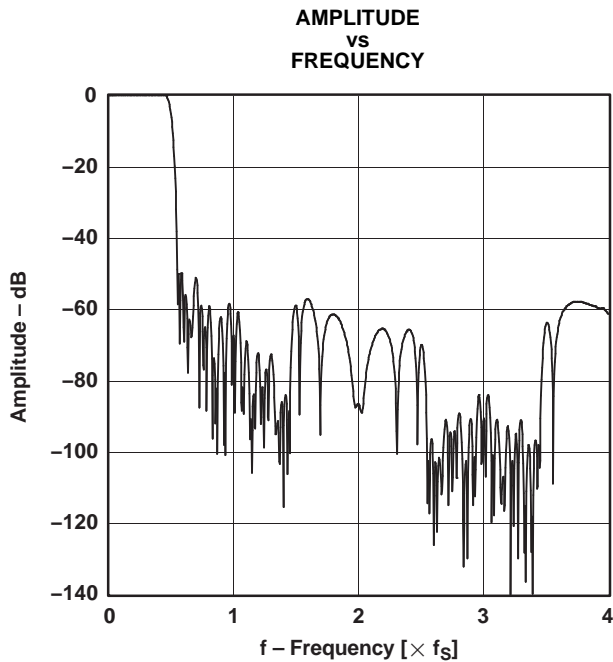
Figure 1.
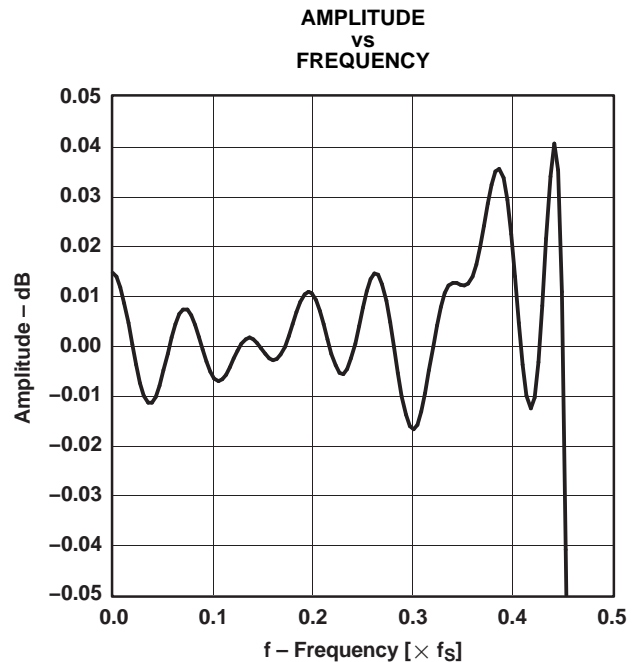
Figure 2.

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TYPICAL PERFORMANCE CURVES (continued)

All specifications at T_A = 25°C, V_{CC} = V_{HP} 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S, 24-bit data, and R_L = 16 Ω, unless otherwise noted.

De-Emphasis Curves
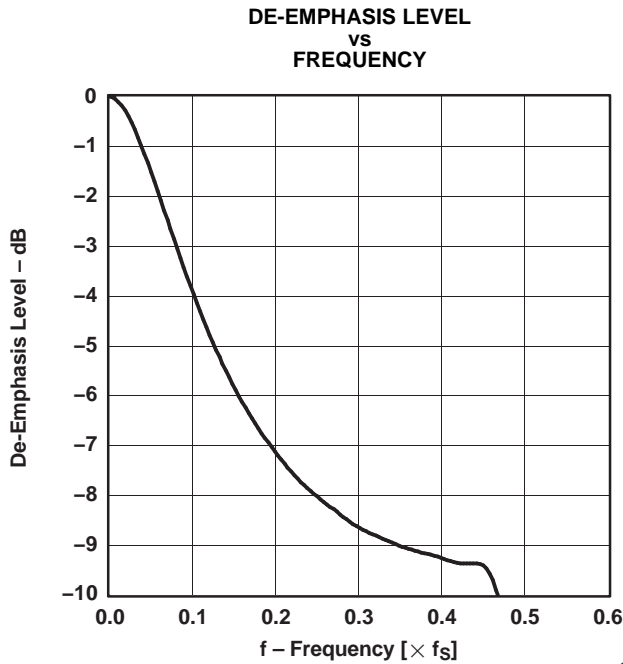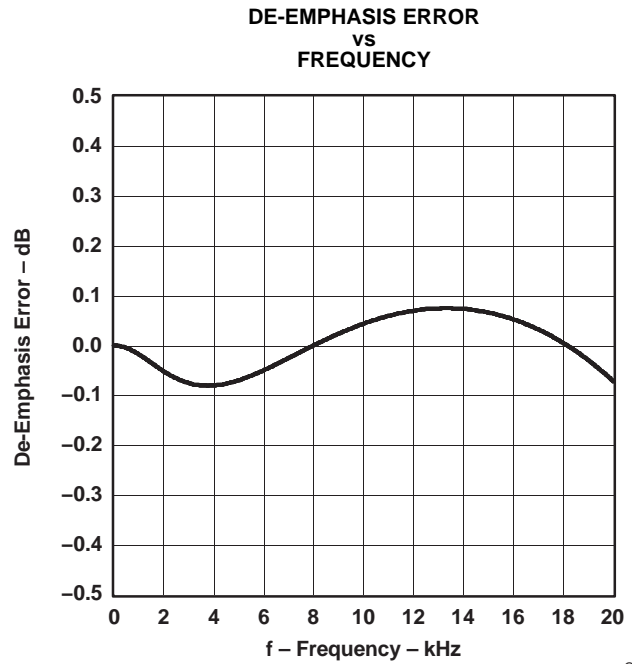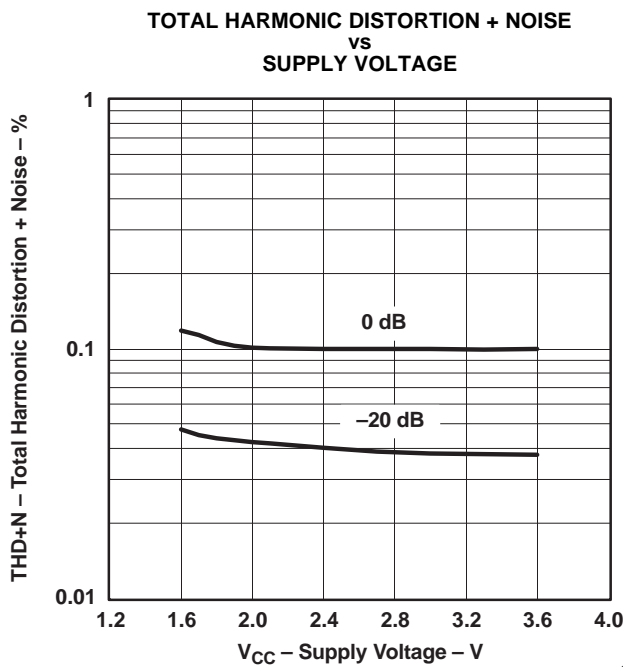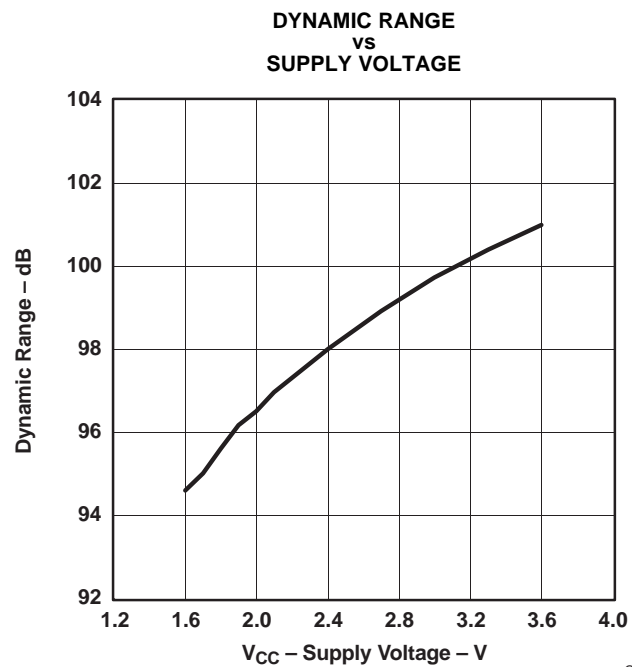
**DE-EMPHASIS LEVEL
vs
FREQUENCY**

Figure 3.

**DE-EMPHASIS ERROR
vs
FREQUENCY**

Figure 4.

**TOTAL HARMONIC DISTORTION + NOISE
vs
SUPPLY VOLTAGE**

Figure 5.

**DYNAMIC RANGE
vs
SUPPLY VOLTAGE**

Figure 6.

TYPICAL PERFORMANCE CURVES (continued)

All specifications at T_A = 25°C, V_{CC} = V_{HP} 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S, 24-bit data, and R_L = 16 Ω, unless otherwise noted.

SIGNAL-TO-NOISE RATIO
vs
SUPPLY VOLTAGE

Figure 7.

CHANNEL SEPARATION
vs
SUPPLY VOLTAGE

Figure 8.

TOTAL HARMONIC DISTORTION + NOISE
vs
FREE-AIR TEMPERATURE

Figure 9.

DYNAMIC RANGE
vs
FREE-AIR TEMPERATURE
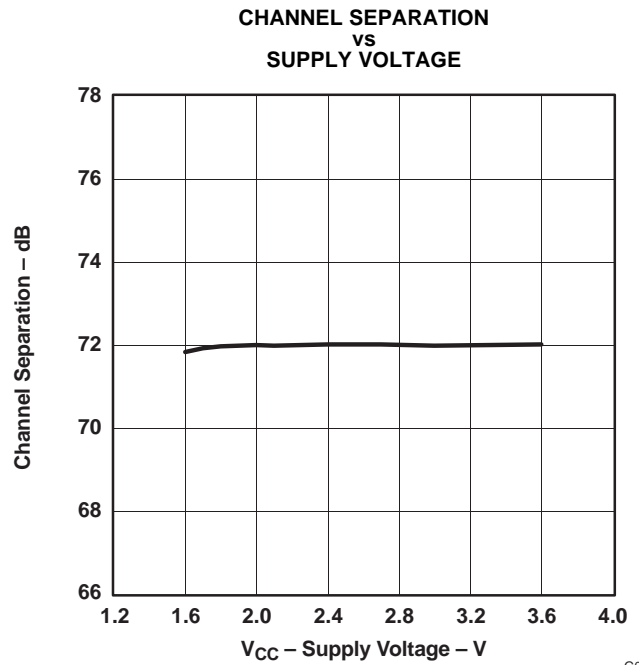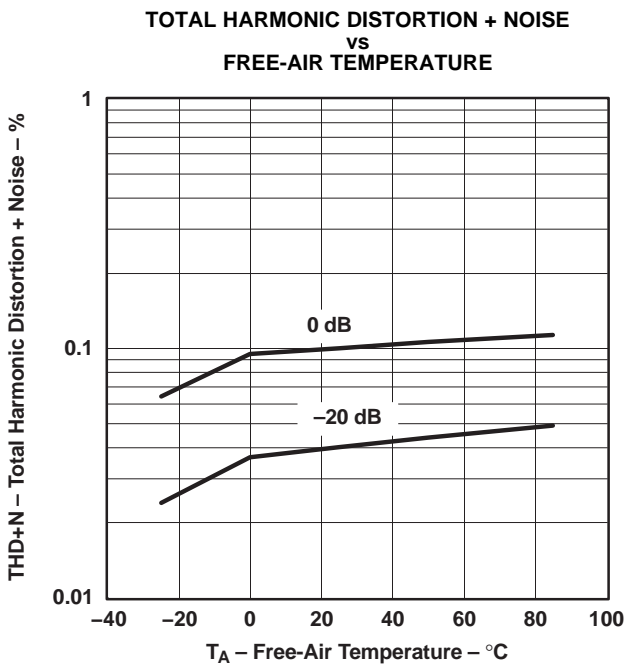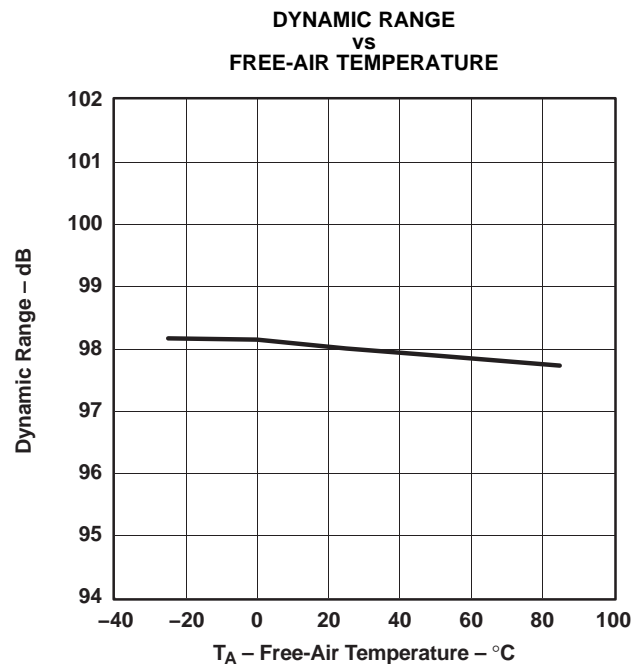
Figure 10.

TYPICAL PERFORMANCE CURVES (continued)

All specifications at T_A = 25°C, V_{CC} = V_{HP} 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S, 24-bit data, and R_L = 16 Ω, unless otherwise noted.

SIGNAL-TO-NOISE RATIO
vs
FREE-AIR TEMPERATURE

Figure 11.

CHANNEL SEPARATION
vs
FREE-AIR TEMPERATURE

Figure 12.

SUPPLY CURRENT
vs
SUPPLY VOLTAGE

Figure 13.

SUPPLY CURRENT
vs
SAMPLING FREQUENCY
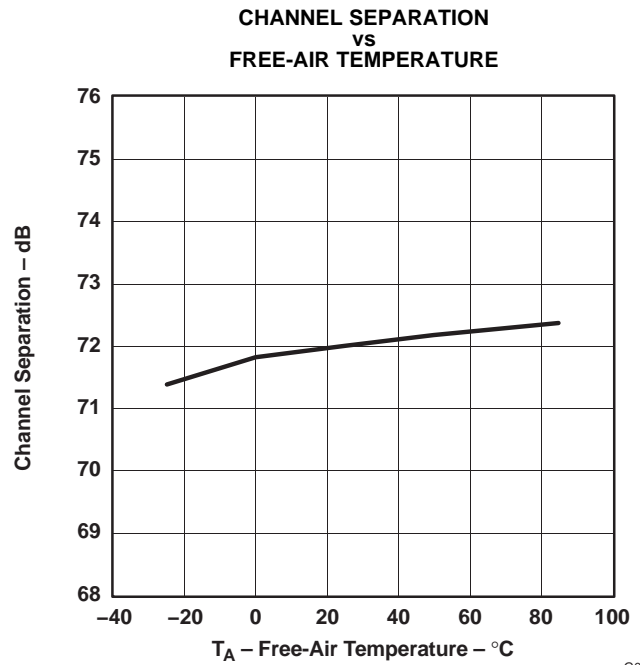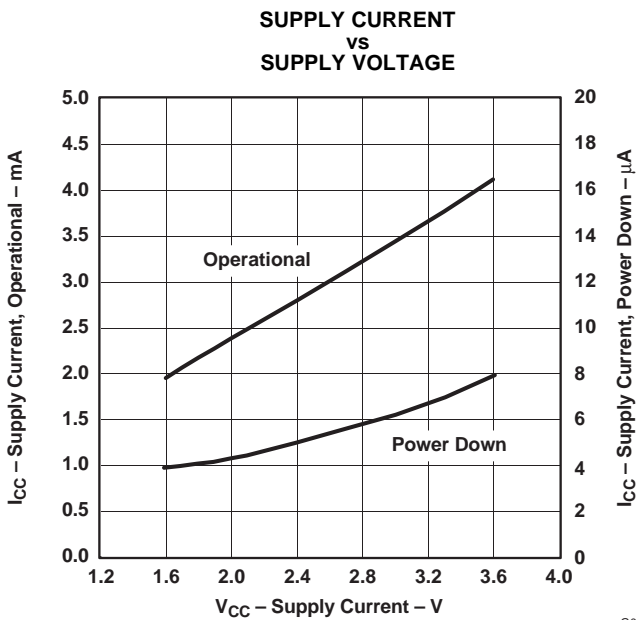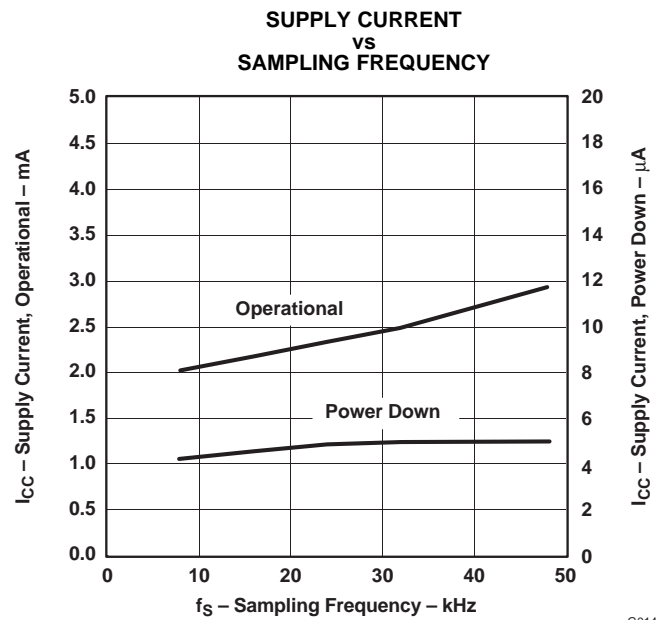
Figure 14.

TYPICAL PERFORMANCE CURVES (continued)

All specifications at T_A = 25°C, V_{CC} = V_{HP} 2.4 V, f_S = 44.1 kHz, system clock = 256 f_S, 24-bit data, and R_L = 16 Ω, unless otherwise noted.

DYNAMIC RANGE vs JITTER

Figure 15.

OUTPUT SPECTRUM (-60 dB, N = 8192)

Figure 16.

OUTPUT SPECTRUM (-60 dB, N = 8192)
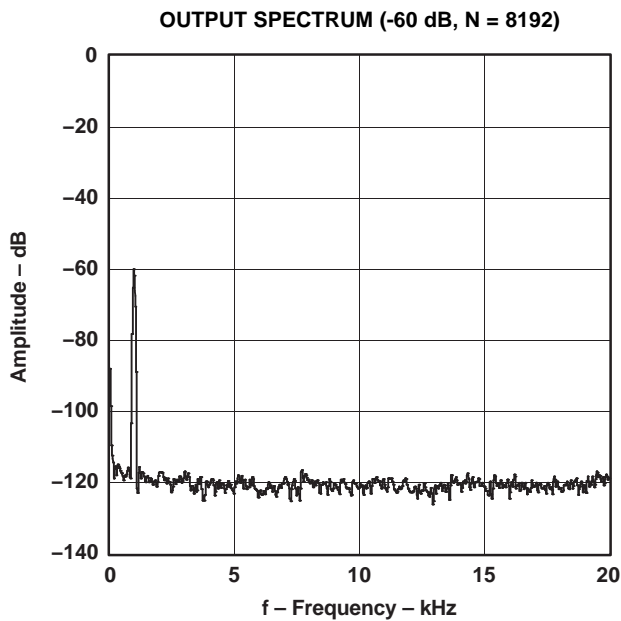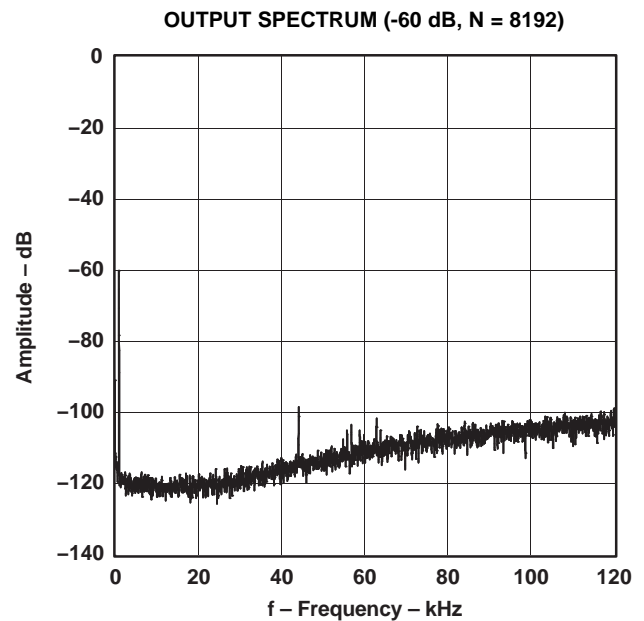
Figure 17.

DETAILED DESCRIPTION

System Clock, Reset, and Functions

System Clock Input

The PCM1770 and PCM1771 devices require a system clock for operating the digital interpolation filters and multilevel Δ-Σ modulators. The system clock is applied at terminal 16 (SCKI). Table 1 shows examples of system clock frequencies for common audio sampling rates.

Figure 18 shows the timing requirements for the system clock input. For optimal performance, it is important to use a clock source with low phase jitter and noise.

Table 1. System Clock Frequency for Common Audio Sampling Frequencies

SAMPLING FREQUENCY, LRCK	SYSTEM CLOCK FREQUENCY, SCKI (MHz)			
	128 f_S	192 f_S	256 f_S	384 f_S
48 kHz	6.144	9.216	12.288	18.432
44.1 kHz	5.6448	8.4672	11.2896	16.9344
32 kHz	4.096	6.144	8.192	12.288
24 kHz	3.072	4.608	6.144	9.216
22.05 kHz	2.8224	4.2336	5.6448	8.4672
16 kHz	2.048	3.072	4.096	6.144
12 kHz	1.536	2.304	3.072	4.608
11.025 kHz	1.4112	2.1168	2.8224	4.2336
8 kHz	1.024	1.536	2.048	3.072
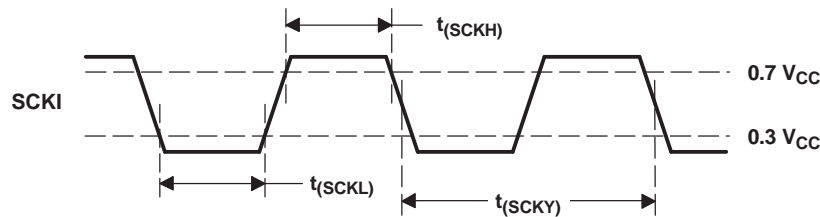
T0005-02

	PARAMETER	MIN	UNIT
$t_{(SCKH)}$	System clock pulse duration, HIGH	7	ns
$t_{(SCKL)}$	System clock pulse duration, LOW	7	ns
$t_{(SCKY)}$	System clock pulse cycle time[1]	52	ns

(1)   1/(128 f_S), 1/(192 f_S), 1/(256 f_S) or 1/(384 f_S)

Figure 18. System Clock Timing

Power On/Off Reset

The PCM1770/71 always must have the $\overline{PD}$ pin set from LOW to HIGH once after power-supply voltages V_{CC} and V_{HP} have reached the specified voltage range and stable clocks SCKI, BCK, and LRCK are being supplied for the power-on sequence. A minimum time of 1 ms after both the clock and power-supply requirements are met is required before the $\overline{PD}$ pin changes from LOW to HIGH, as shown in Figure 19. Subsequent to the $\overline{PD}$ LOW-to-HIGH transition, the internal logic state is held in reset for 1024 system clock cycles prior to the start of the power-on sequence. During the power-on sequence, $H_{OUT}L$ and $H_{OUT}R$ increase gradually from ground leved, reaching an output level that corresponds to the input data after a period of $9334/f_S$. When powering off, the $\overline{PD}$ pin is set from HIGH to LOW first. Then $H_{OUT}L$ and $H_{OUT}R$ decrease gradually to ground level over a period of $9334/f_S$, as shown in Figure 20, after which power can be removed without creating pop noise. When powering on or off, adhering to the timing requirements of Figure 19 and Figure 20 ensures that pop noise does not occur. If the timing requirements are not met, pop noise might occur.
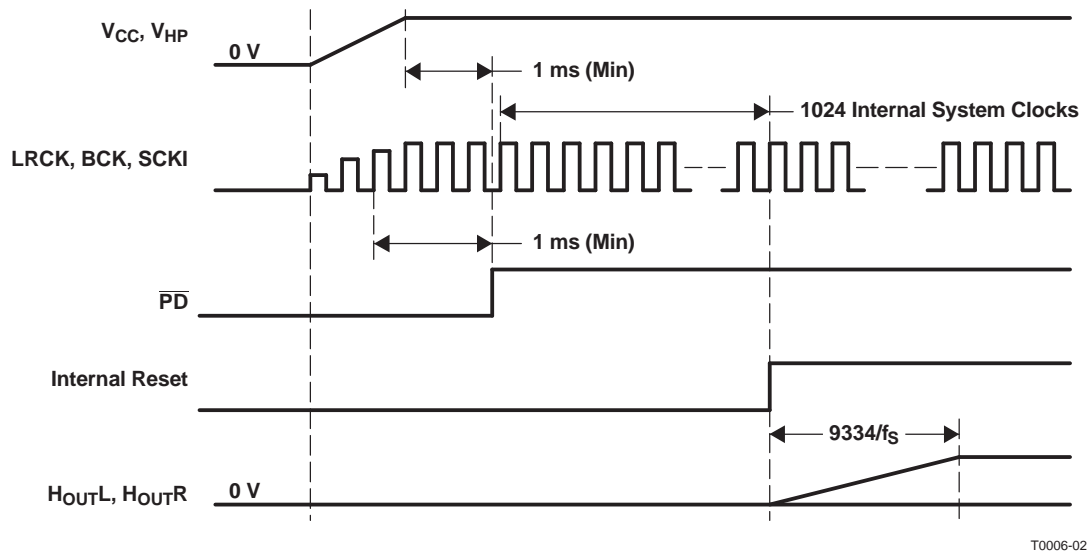
Figure 19. Power-On Sequence
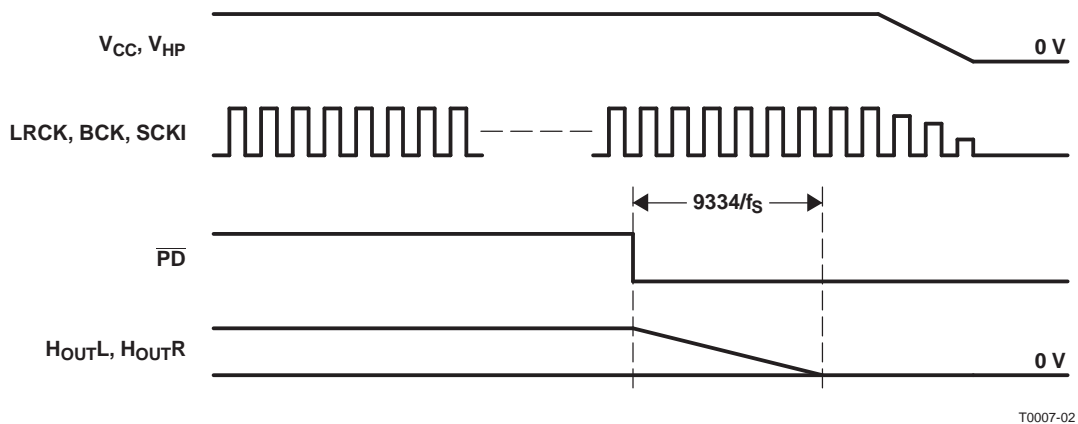
Figure 20. Power-Off Sequence

Power-Up/-Down Sequence and Reset

The PCM1770 device has two kinds of power-up/-down methods: the $\overline{PD}$ terminal through hardware control and PWRD (register 4, B0) through software control. The PCM1771 device has only the $\overline{PD}$ terminal through hardware control for the power-up/-down sequence. The power-up or power-down sequence operates the same as the power-on or power-off sequence. When powering up or down using the $\overline{PD}$ terminal, all digital circuits are reset. When powering up or down using PWRD, all digital circuits are reset except for maintaining the logic states of the registers. Figure 21 shows the power-up/power-down sequence.
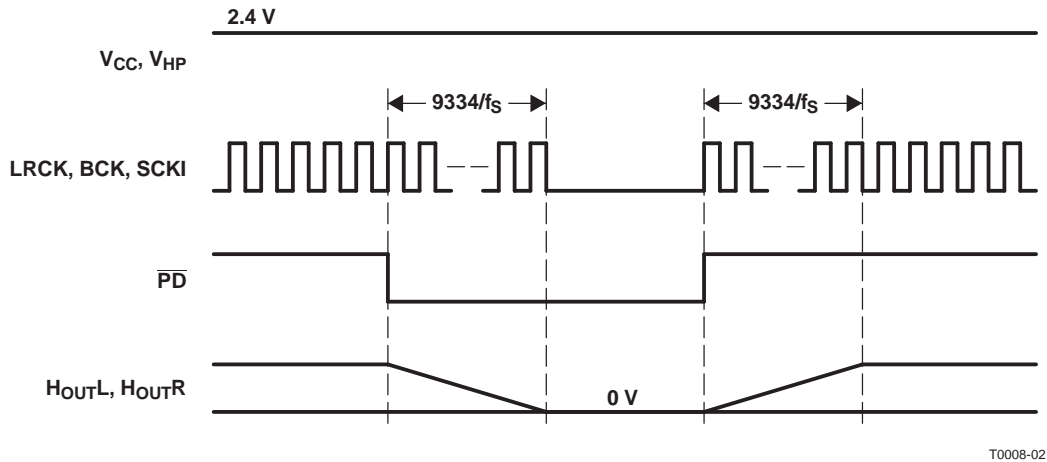
Figure 21. Power-Down and Power-Up Sequences

Audio Serial Interface

The audio serial interface for the PCM1770 and PCM1771 devices consists of a 3-wire synchronous serial port. It includes terminals 1 (LRCK), 2 (DATA), and 3 (BCK). BCK is the serial audio bit clock, and it clocks the serial data present on DATA into the audio interface serial shift register. Serial data is clocked into the PCM1770 and PCM1771 devices on the rising edge of BCK. LRCK is the serial audio left/right word clock. It latches serial data into the serial audio interface internal registers.

Both LRCK and BCK of the PCM1770 device support the slave and master modes, which are set by FMT (register 3). LRCK and BCK are outputs during the master mode and inputs during the slave mode.

In slave mode, BCK and LRCK are synchronous to the audio system clock, SCKI. Ideally, it is recommended that LRCK and BCK be derived from SCKI. LRCK is operated at the sampling frequency, f_S. BCK can be operated at 32, 48, or 64 times the sampling frequency.

In master mode, BCK and LRCK are derived from the system clock and these terminals are outputs. BCK and LRCK are synchronous to SCKI. LRCK is operated at the sampling frequency, f_S. BCK can be operated at 64 times the sampling frequency.

The PCM1770 and PCM1771 devices operate under LRCK synchronized with the system clock. The PCM1770 and PCM1771 devices do not need a specific phase relationship between LRCK and the system clock, but do require the synchronization of LRCK and the system clock. If the relationship between the system clock and LRCK changes more than ±3 BCK during one sample period, internal operation of the PCM1770 and PCM1771 devices halts within $1/f_S$, and the analog output is kept in last data until resynchronization between system clock and LRCK is completed.

Audio Data Formats and Timing

The PCM1770 device supports industry-standard audio data formats, including standard, I^{2}S, and left-justified. The PCM1771 device supports the I^{2}S and left-justified data formats. Table 2 lists the main features of the audio data interface. Figure 22 shows the data formats. Data formats are selected using the format bits, FMT[2:0] of control register 3 in the case of the PCM1770 device, and are selected using the FMT terminal in the case of the PCM1771 device. The default data format is 24-bit, left-justified, slave mode. All formats require binary 2s complement, MSB-first audio data. Figure 23 shows a detailed timing diagram for the serial audio interface in slave mode. Figure 24 shows a detailed timing diagram for the serial audio interface in master mode.

**Table 2. Audio Data Interface**

AUDIO-DATA INTERFACE FEATURE		CHARACTERISTIC
Audio data interface format	(PCM1770)	Standard, I^{2}S, left-justified
	(PCM1771)	I^{2}S, left-justified
Audio data bit length		16-, 20-, 24-bit selectable
Audio data format		MSB-first, 2s-complement

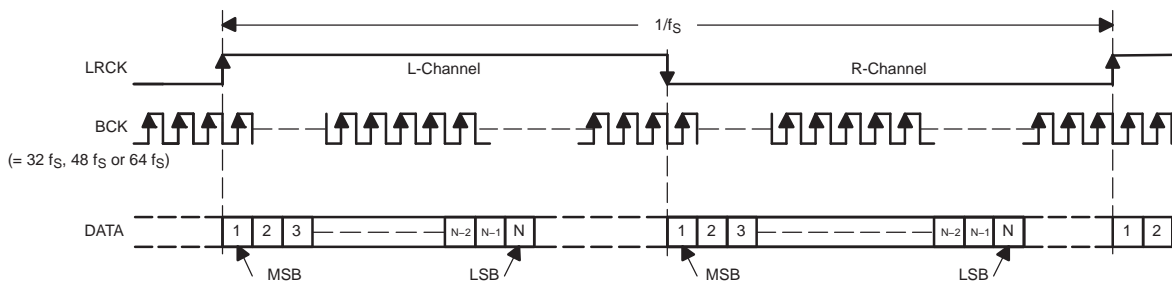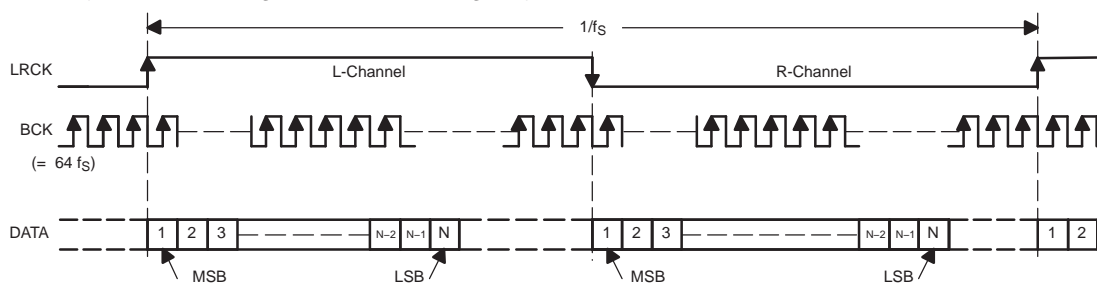**(1) Standard Data Format; L-Channel = HIGH, R-Channel = LOW (Slave Mode)**
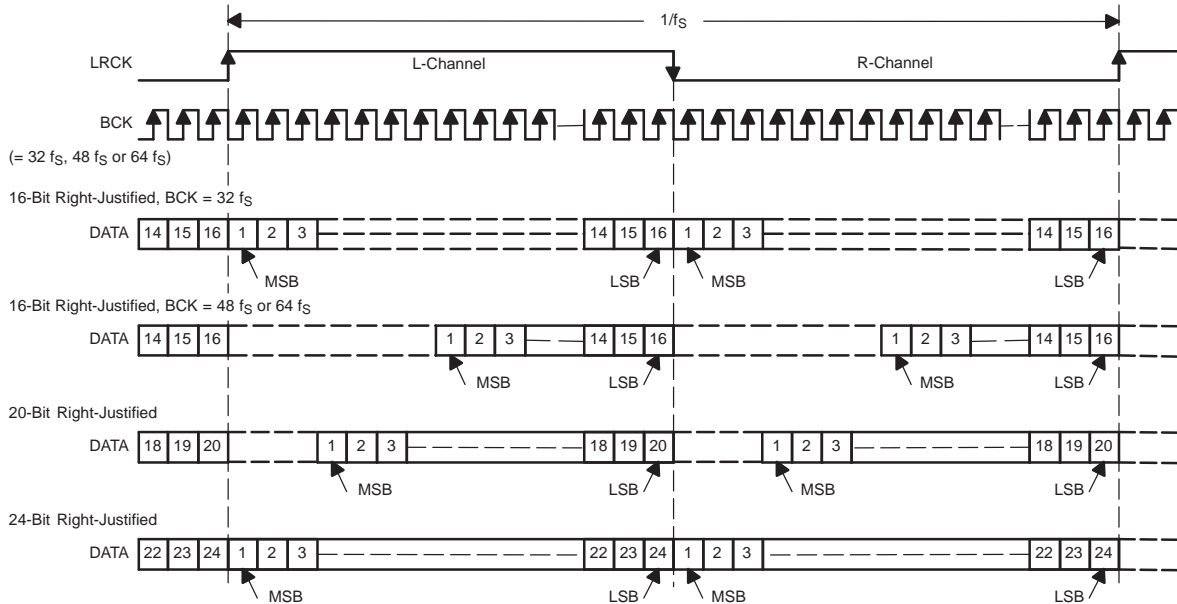
(2) I²S Data Format; L-Channel = LOW, R-Channel = HIGH (Slave Mode)
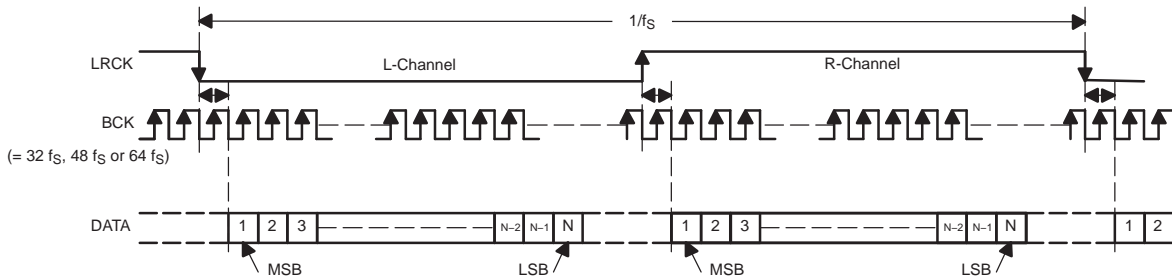
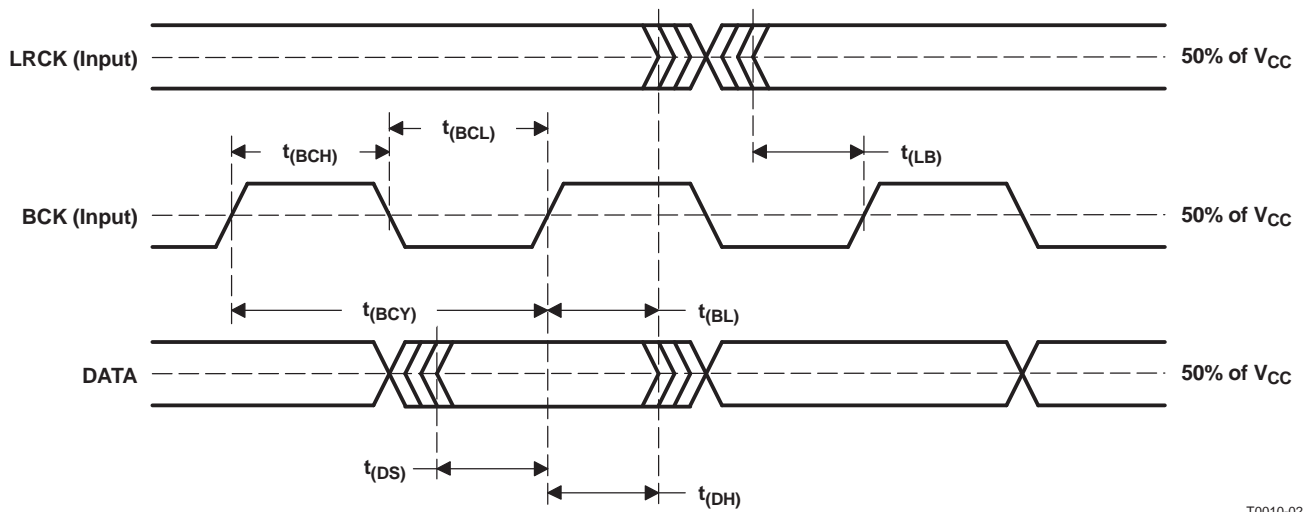**(3) Left-Justified Data Format; L-Channel = HIGH, R-Channel = LOW (Slave Mode)**

(4) Left-Justified Data Format; L-Channel = HIGH, R-Channel = LOW (Master Mode)
(The frequency of BCK is 64 f_S and SCKI is 256 f_S only)

T0009-01
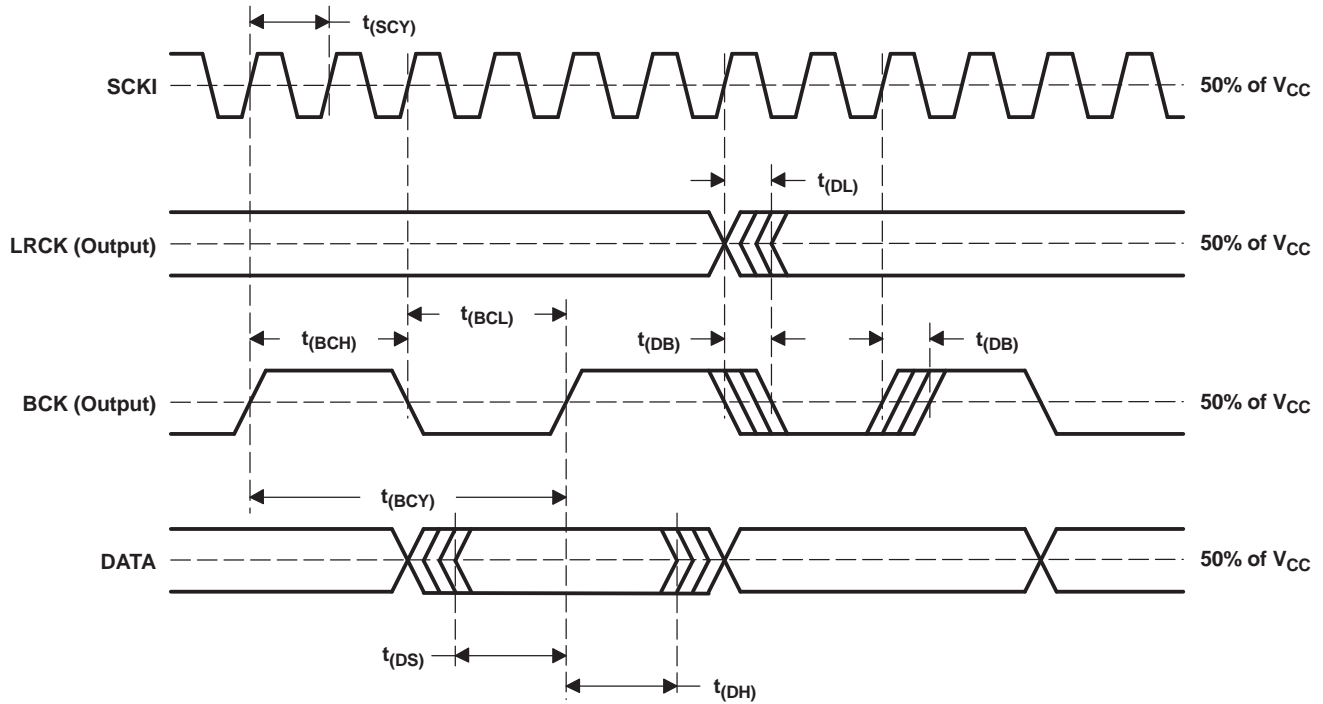
Figure 22. Audio Data Input Formats

Figure 23. Audio Interface Timing (Slave Mode)

PARAMETER		MIN	MAX	UNIT
$t_{(BCY)}$	BCK pulse cycle time	$1/(64\ f_S)$[1]		
$t_{(BCH)}$	BCK high-level time	35		ns
$t_{(BCL)}$	BCK low-level time	35		ns
$t_{(BL)}$	BCK rising edge to LRCK edge	10		ns
$t_{(LB)}$	LRCK edge to BCK rising edge	10		ns
$t_{(DS)}$	DATA set-up time	10		ns
$t_{(DH)}$	DATA hold time	10		ns

(1) f_S is the sampling frequency.

T0011-02

PARAMETER		MIN	MAX	UNIT
$t_{(SCY)}$	SCKI pulse cycle time	$1/(256 \, f_S)$[1]		
$t_{(DL)}$	LRCK edge from SCKI rising edge	0	40	ns
$t_{(DB)}$	BCK edge from SCKI rising edge	0	40	ns
$t_{(BCY)}$	BCK pulse cycle time	$1/(64 \, f_S)$[1]		
$t_{(BCH)}$	BCK high-level time	146		ns
$t_{(BCL)}$	BCK low-level time	146		ns
$t_{(DS)}$	DATA setup time	10		ns
$t_{(DH)}$	DATA hold time	10		ns

(1) f_S is up to 48 kHz. f_S is the sampling frequency.

Figure 24. Audio Interface Timing (Master Mode)

Hardware Control (PCM1771)

The digital functions of the PCM1771 device are capable of hardware control. Table 3 shows selectable formats, Table 4 shows de-emphasis control, and Table 5 shows analog mixing control.

Table 3. Data Format Select

FMT	DATA FORMAT
Low	16- to 24-bit, left-justified format
High	16- to 24-bit, I^{2}S format

Table 4. De-Emphasis Control

DEMP	DE-EMPHASIS FUNCTION
Low	44.1-kHz de-emphasis OFF
High	44.1-kHz de-emphasis ON

Table 5. Analog Mixing Control

AMIX	ANALOG MIXING
Low	Analog mixing OFF
High	Analog mixing ON

Software Control (PCM1770)

The PCM1770 device has many programmable functions that can be controlled in the software-control mode. The functions are controlled by programming the internal registers using $\overline{MS}$, MC, and MD.

The software-control interface is a 3-wire serial port that operates asynchronously to the serial audio interface. The serial control interface is used to program the on-chip mode registers. MD is the serial data input, used to program the mode registers. MC is the serial bit clock, used to shift data into the control port. $\overline{MS}$ is the mode control port select signal.

Register Write Operation (PCM1770)

All write operations for the serial control port use 16-bit data words. Figure 25 shows the control data word format. The most significant bit must be 0. There are seven bits, labeled IDX[6:0], that set the register index (or address) for the write operation. The eight least significant bits, D[7:0], contain the data to be written to the register specified by IDX[6:0].

Figure 26 shows the functional timing diagram for writing to the serial control port. To write data into the mode register, the data is clocked into an internal shift register on the rising edge of the MC clock. Serial data can change on the falling edge of the MC clock and must be stable on the rising edge of the MC clock. The $\overline{MS}$ signal must be low during the write mode and the rising edge of the $\overline{MS}$ signal must be aligned with the falling edge of the last MC clock pulse in the 16-bit frame. The MC clock can run continuously between transactions while the $\overline{MS}$ signal is low.

MSB															LSB
0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	D7	D6	D5	D4	D3	D2	D1	D0

Register Index (or Address)   Register Data
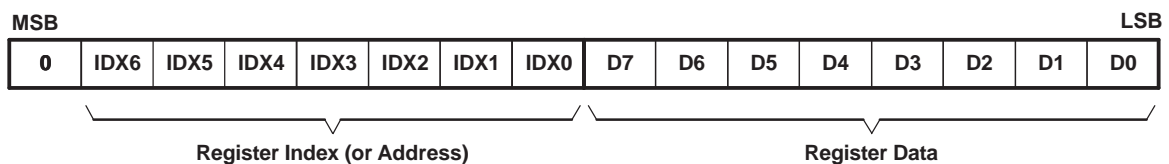
R0001-01

Figure 25. Control Data Word Format for MD
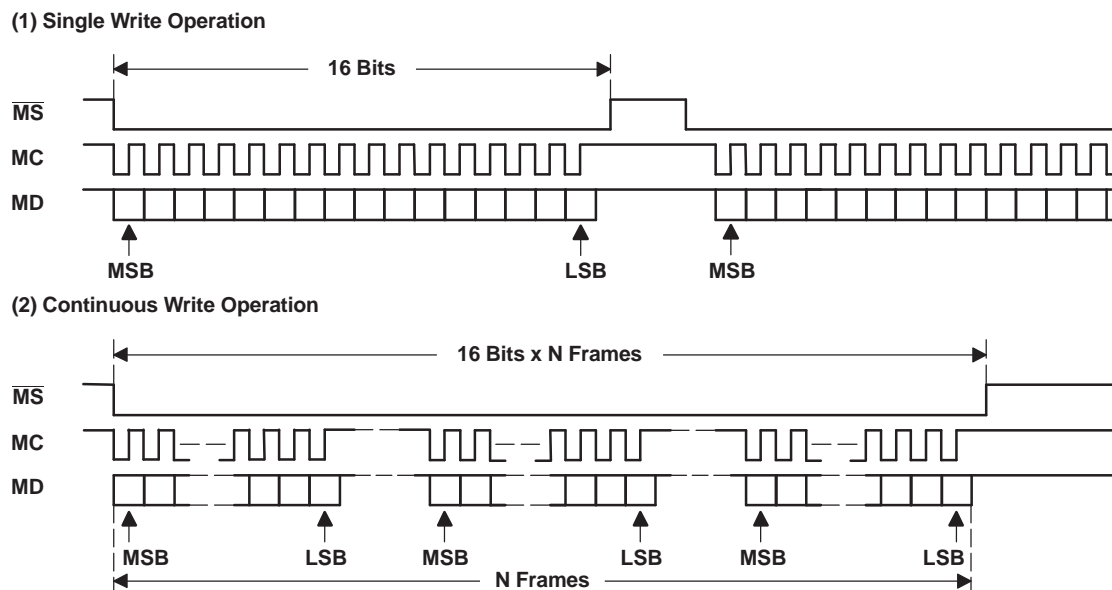
Figure 26. Register Write Operation

Control Interface Timing Requirements (PCM1770)

Figure 27 shows a detailed timing diagram for the serial control interface. These timing parameters are critical for proper control port operation.
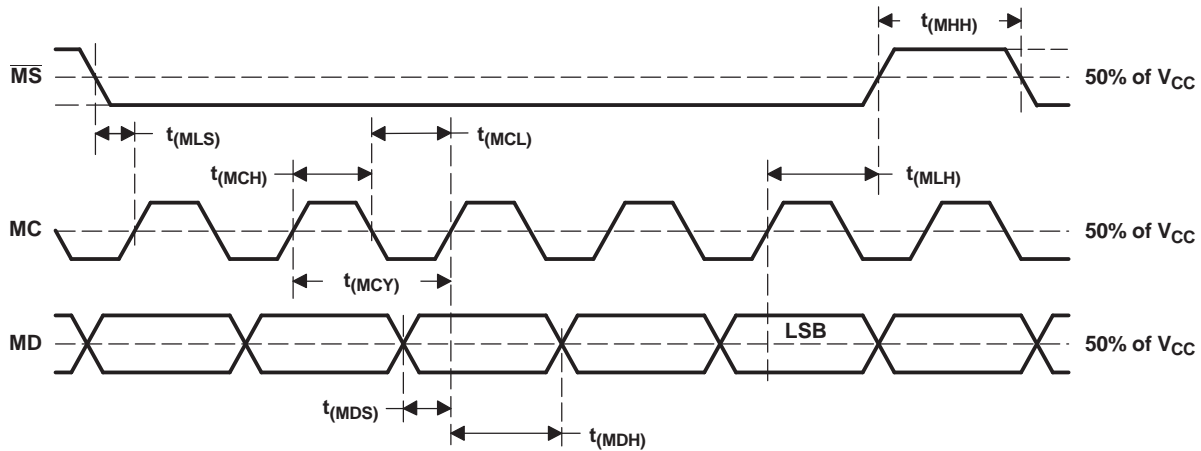
T0013-02

PARAMETER	SYMBOL	MIN	TYP	MAX	UNIT
MC pulse cycle time	$t_{(MCY)}$	100[1]			ns
MC low-level time	$t_{(MCL)}$	50			ns
MC high-level time	$t_{(MCH)}$	50			ns
$\overline{MS}$ high-level time	$t_{(MHH)}$	[2]			ns
$\overline{MS}$ falling edge to MC rising edge	$t_{(MLS)}$	20			ns
$\overline{MS}$ hold time	$t_{(MLH)}$	20			ns
MD hold time	$t_{(MDH)}$	15			ns
MD setup time	$t_{(MDS)}$	20			ns

(1) When MC runs continuously between transactions, MC pulse cycle time is specified as $3/(128\,f_S)$, where f_S is sampling rate.

(2) $3/(128\,f_S)$ s (min), where f_S is sampling rate.

Figure 27. Control Interface Timing

Mode Control Registers (PCM1770)

User-Programmable Mode Controls

The PCM1770 device has a number of user-programmable functions that can be accessed via mode control registers. The registers are programmed using the serial control interface, as discussed in the *Software Control (PCM1770)* section. Table 6 lists the available mode control functions, along with their reset default conditions and associated register index.

Register Map

Table 7 shows the mode control register map. Each register includes an index (or address) indicated by the IDX[6:0] bits.

Table 6. User-Programmable Mode Controls

FUNCTION	RESET DEFAULT	REGISTER NO.	BIT(S)
Soft mute control, L/R independently	Disabled	01	MUTL, MUTR
Digital attenuation level setting, 0 dB to –63 dB in 1-dB steps, L/R independently	0 dB	01, 02	ATL[5:0], ATR[5:0]
Oversampling rate control (128 f_S, 192 f_S, 256 f_S, 384 f_S)	128 f_S oversampling	03	OVER
Polarity control for analog output for R-channel DAC	Not inverted	03	RINV
Analog mixing control for analog in, AIN (terminal 14)	Disabled	03	AMIX
44.1-kHz de-emphasis control	Disabled	03	DEM
Audio data format select	24-bit, left-justified format	03	FMT[2:0]
Zero cross attenuation	Disabled	04	ZCAT
Power-down control	Disabled	04	PWRD

Table 7. Mode Control Register Map

REGISTER	IDX [6:0] (B14-B8)	B15	B14	B13	B12	B11	B10	B9	B8	B7	B6	B5	B4	B3	B2	B1	B0
01	01h	0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	MUTR	MUTL	ATL5	ATL4	ATL3	ATL2	ATL1	ATL0
02	02h	0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	RSV (1)	RSV (1)	ATR5	ATR4	ATR3	ATR2	ATR1	ATR0
03	03h	0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	OVER	RSV (1)	RINV	AMIX	DEM	FMT2	FMT1	FMT0
04	04h	0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	RSV (1)	RSV (1)	RSV (1)	ZCAT	RSV (1)	RSV (1)	RSV (1)	PWR

(1) RSV is reserved for test operation. It must be set to 0 during regular operation.

Register Definitions

Register 01

B15	B14	B13	B12	B11	B10	B9	B8	B7	B6	B5	B4	B3	B2	B1	B0
0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	MUTR	MUTL	ATL5	ATL4	ATL3	ATL2	ATL1	ATL0

IDX[6:0]: 000 0001b

MUTx: Soft Mute Control

Where, x = L or R, corresponding to the headphone output $H_{OUT}L$ or $H_{OUT}R$.

Default value: 0

MUTL, MUTR = 0	Mute disabled (default)
MUTL, MUTR = 1	Mute enabled

The mute bits, MUTL and MUTR, enable or disable the soft mute function for the corresponding headphone outputs, $H_{OUT}L$ and $H_{OUT}R$. The soft mute function is incorporated into the digital attenuators. When mute is disabled (MUTx = 0), the attenuator and DAC operate normally. When mute is enabled by setting MUTx = 1, the digital attenuator for the corresponding output is decreased from the current setting to the infinite attenuation, one attenuator step (1 dB) at a time. This provides pop-free muting of the headphone output.

By setting MUTx = 0, the attenuator is increased one step at a time to the previously programmed attenuation level.

ATL[5:0]: Digital Attenuation Level Setting for Headphone Output, $H_{OUT}L$

Default value: 11 1111b

Headphone output $H_{OUT}L$ includes a digital attenuation function. The attenuation level can be set from 0 dB to −62 dB, in 1-dB steps. Changes in attenuator levels are made by incrementing or decrementing by one step (1 dB) for every $8/f_S$ time internal until the programmed attenuator setting is reached. Alternatively, the attenuation level may be set to infinite attenuation (or mute).

The following table shows the attenuation levels for various settings:

ATL[5:0]	ATTENUATION LEVEL SETTING
11 1111b	0 dB, no attenuation (default)
11 1110b	−1 dB
11 1101b	−2 dB
:	:
00 0010b	−61 dB
00 0001b	−62 dB
00 0000b	Mute

Register 02

B15	B14	B13	B12	B11	B10	B9	B8	B7	B6	B5	B4	B3	B2	B1	B0
0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	RSV	RSV	ATR5	ATR4	ATR3	ATR2	ATR1	ATR0

IDX[6:0]: 000 0010b

ATR[5:0]: Digital Attenuation Level Setting for Headphone Output, $H_{OUT}R$

Default value: 11 1111b

Headphone output $H_{OUT}R$ includes a digital attenuation function. The attenuation level can be set from 0 dB to −62 dB, in 1-dB steps. Changes in attenuator levels are made by incrementing or decrementing by one step (1 dB) for every $8/f_S$ time internal until the programmed attenuator setting in reached. Alternatively, the attenuation level can be set to infinite attenuation (or mute).

To set the attenuation levels for ATR[5:0], see the table for ATL[5:0], register 01.

Register 03

B15	B14	B13	B12	B11	B10	B9	B8	B7	B6	B5	B4	B3	B2	B1	B0
0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	OVER	RSV	RINV	AMIX	DEM	FMT2	FMT1	FMT0

IDX[6:0]: 000 0011b

OVER: Oversampling Control

Default value: 0

OVER = 0	$128f_S$ oversampling
OVER = 1	$192f_S$, $256f_S$, $384f_S$ oversampling

The OVER bit controls the oversampling rate of the $\Delta\text{-}\Sigma$ D/A converters. When it operates at a low sampling rate, less than 24 kHz, this function is recommended.

RINV: Polarity Control for Headphone Output, $H_{OUT}R$

Default value: 0

RINV = 0	Not inverted
RINV = 1	Inverted output

The RINV bits allow the user to control the polarity of the headphone output, $H_{OUT}R$. This function can be used to connect the monaural speaker using the BTL connection method. This bit is recommended to be 0 during the power-up/-down sequence for minimizing audible pop noise.

AMIX: Analog Mixing Control for External Analog Signal, AIN

Default value: 0

AMIX = 0	Disabled (not mixed)
AMIX = 1	Enabled (mixing to the DAC output)

The AMIX bit allows the user to mix analog input (AIN) with headphone outputs ($H_{OUT}L/H_{OUT}R$) internally.

DEM: 44.1-kHz De-Emphasis Control

Default value: 0

DEM = 0	Disabled
DEM = 1	Enabled

The DEM bit enables or disables the digital de-emphasis filter for the 44.1-kHz sampling rate.

FMT[2:0]: Audio Interface Data Format

Default value: 000

The FMT[2:0] bits select the data format for the serial audio interface. The following table shows the available format options.

FMT[2:0]	Audio Data Format Selection
000	16- to 24-bit, left-justified format (default)
001	16- to 24-bit, I^2S format
010	24-bit right-justified data
011	20-bit right-justified data
100	16-bit right-justified data
101	16- to 24-bit, left-justified format, master mode
110	Reserved
111	Reserved

Register 04

B15	B14	B13	B12	B11	B10	B9	B8	B7	B6	B5	B4	B3	B2	B1	B0
0	IDX6	IDX5	IDX4	IDX3	IDX2	IDX1	IDX0	RSV	RSV	RSV	ZCAT	RSV	RSV	RSV	PWRD

IDX[6:0]: 0000 0100b

ZCAT: Zero Cross Attenuation

Default value: 0

ZCAT = 0	Normal attenuation (default)
ZCAT = 1	Zero cross attenuation

This bit enables changing the signal level on zero crossing during attenuation control or muting. If the signal does not cross BPZ beyond $512/f_S$ (11.6 ms at 44.1-kHz sampling rate), the signal level is changed similarly to normal attenuation control. This function is independently monitored for each channel; moreover, change of signal level is alternated between both channels. Figure 28 shows an example of zero cross attenuation.

ATT CTRL START

L-Channel
(1.5 kHz)

R-Channel
(1 kHz)

○ **Level Change Point**

W0001-01
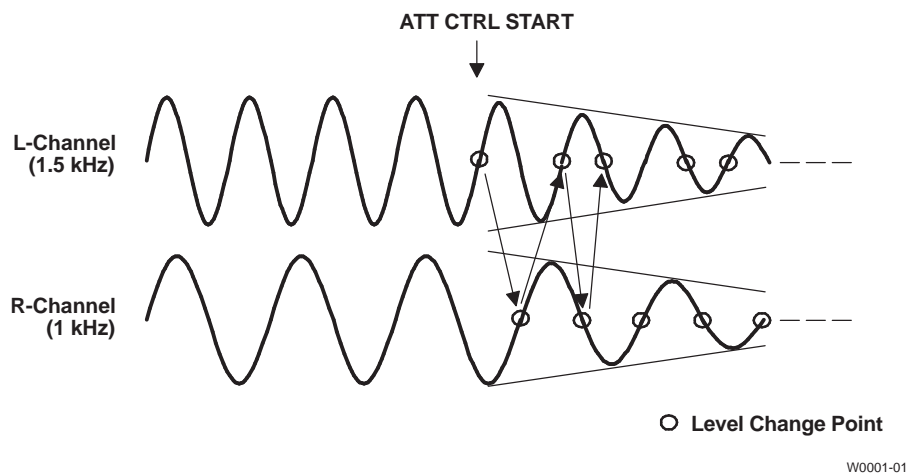
Figure 28. Example of Zero Cross Attenuation

PWRD: Power Down Control

Default value: 0

PWRD = 0	Normal operation (default)
PWRD = 1	Power-down state

This bit is used to enter into low-power mode. Note that PWRD has no reset function.

When this bit is set to 1, the PCM1770 device enters low-power mode and all digital circuits are reset except the register states, which remain unchanged.

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Analog In/Out

Headphone Output (Stereo)

The PCM1770 and PCM1771 devices have two independent headphone amplifiers, and the amplifier outputs are provided at the $H_{OUT}L$ and $H_{OUT}R$ terminals. Because the capability of the headphone output is designed for driving a 16-Ω impedance headphone, less than a 16-Ω impedance headphone is not recommended. A resistor and a capacitor must be connected to $H_{OUT}L$ and $H_{OUT}R$ to ensure proper output loading.

Monaural Output (BTL Mode/Monaural Speaker)

The monaural output can be created by summing the left and right headphone outputs. When in the BTL mode, the user must set the headphone output levels to –3 dB using the ATL[5:0] bits in register 01 and the ATR[5:0] bits in register 02. Moreover, invert the polarity of the right headphone output by using the RINV bit on control register 03. The RINV bit is recommended to be 0 during the power-up/-down sequence for minimizing audible pop noise.

Analog Input

The PCM1770 and PCM1771 devices have an analog input, AIN (terminal 10). The AMIX bit (PCM1770) or the AMIX terminal (PCM1771) allows the user to mix AIN with the headphone outputs ($H_{OUT}L$ and $H_{OUT}R$) internally. When in the mixing mode, an ac-coupling capacitor is needed for AIN. But if AIN is not used, AIN must be open and the AMIX bit (PCM1770) must be disabled or the AMIX terminal (PCM1771) must be low.

Because AIN does not have an internal low-pass filter, it is recommended that the bandwidth of the input signal into AIN is limited to less than 100 kHz. The source of signals connected to AIN must be connected by low impedance.

Although the maximum input voltage on AIN is designed to be as large as 0.584 V_{HP} (peak-to-peak), the user must attenuate the input voltage on AIN and control the digital input data so that each line output ($H_{OUT}L$ and $H_{OUT}R$) does not exceed 0.55 V_{HP} (peak-to-peak) in the mixing mode.

V_{COM} Output

One unbuffered common-mode voltage output terminal, V_{COM}, is brought out for decoupling purposes. This terminal is nominally biased to a dc voltage level equal to 0.5 V_{HP} and connected to a 10-μF capacitor. In the case of a capacitor smaller than 10 μF, pop noise can be generated during the power-on/-off or power-up/-down sequences.

APPLICATION INFORMATION

Connection Diagrams

Figure 29 shows the basic connection diagram with the necessary power supply bypassing and decoupling components. It is recommended that the component values shown in Figure 29 be used for all designs.

The use of series resistors (22 Ω to 100 Ω) is recommended for the SCKI, LRCK, BCK, and DATA inputs. The series resistor combines with the stray PCB and device input capacitance to form a low-pass filter that reduces high-frequency noise emissions and helps to dampen glitches and ringing present on the clock and data lines.

Power Supplies and Grounding

The PCM1770 and PCM1771 devices require a 2.4-V typical analog supply for V_{CC} and V_{HP}. These 2.4-V supplies power the DAC, analog output filter, and other circuits. For best performance, these 2.4-V supplies must be derived from the analog supply using a linear regulator, as shown in Figure 29.

Figure 29 shows the proper power supply bypassing. The 10-µF capacitors must be tantalum or aluminum electrolytic, while the 0.1-µF capacitors are ceramic (X7R type is recommended for surface-mount applications).

Short-Circuit Protection

Continuous shorting of $H_{OUT}L$ and $H_{OUT}R$ to GND, to a power supply, or to each other is not permitted, as protection circuitry for an output short is not implemented in the device. If the possibility of shorting cannot be eliminated in an application, an 8-Ω or higher series resistor must be added between the phase compensation circuits of the $H_{OUT}x$ pins and the application circuitry (headphone jack in Figure 29).

APPLICATION INFORMATION (continued)
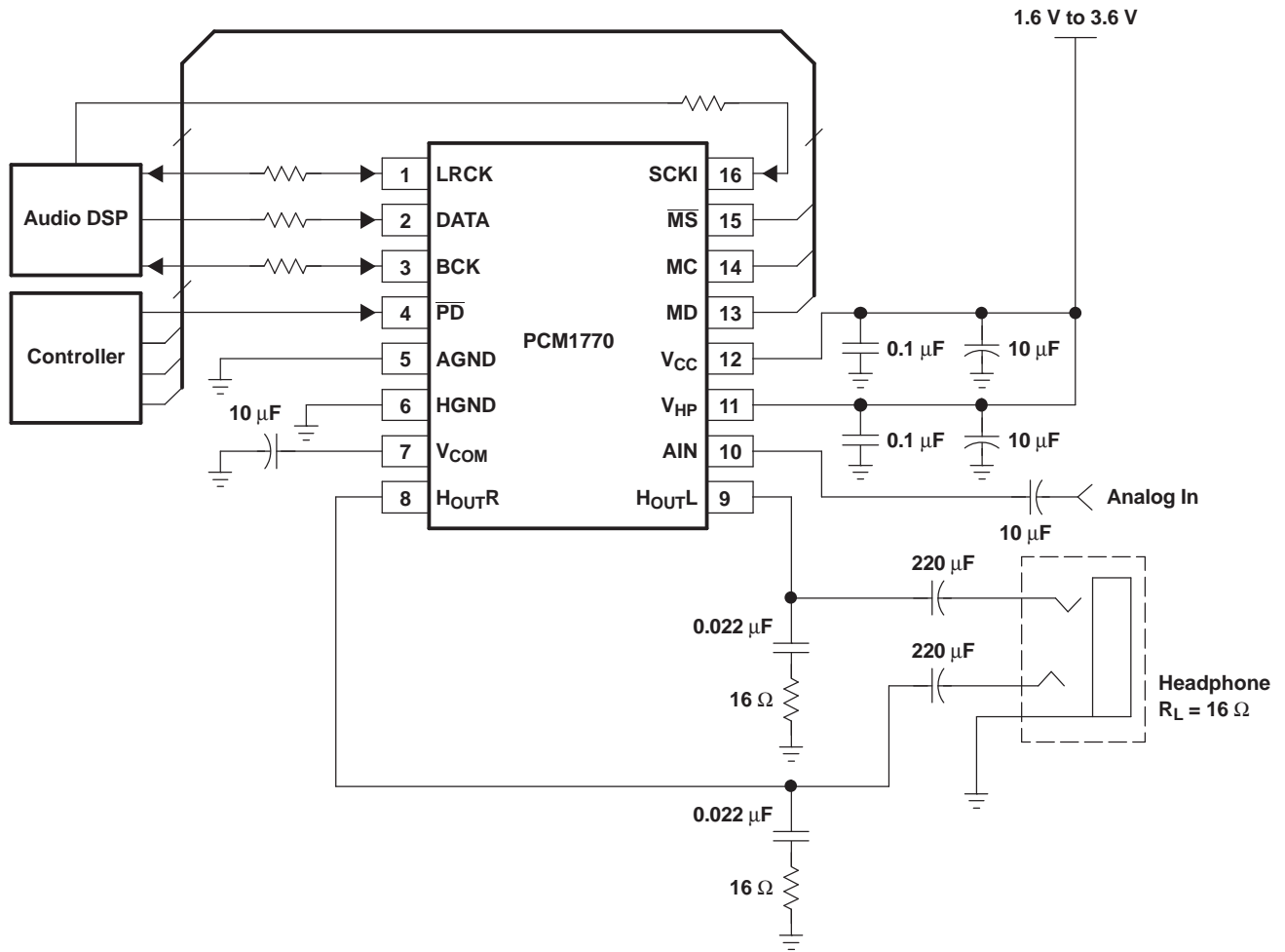
S0008-01

Figure 29. Basic Connection Diagram

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Revision History

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

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
PCM1770PW	ACTIVE	TSSOP	PW	16	250	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1770	Samples
PCM1770PWG4	ACTIVE	TSSOP	PW	16	250	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1770	Samples
PCM1770PWR	ACTIVE	TSSOP	PW	16	2000	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1770	Samples
PCM1770RGA	ACTIVE	VQFN	RGA	20	250	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1770	Samples
PCM1770RGAR	ACTIVE	VQFN	RGA	20	2000	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1770	Samples
PCM1771PW	ACTIVE	TSSOP	PW	16	250	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1771	Samples
PCM1771PWG4	ACTIVE	TSSOP	PW	16	250	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1771	Samples
PCM1771PWR	ACTIVE	TSSOP	PW	16	2000	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1771	Samples
PCM1771PWRG4	ACTIVE	TSSOP	PW	16	2000	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM	0 to 70	1771	Samples
PCM1771RGA	ACTIVE	VQFN	RGA	20	250	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-1-260C-UNLIM		1771	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.

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PACKAGE MATERIALS INFORMATION

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

User Direction of Feed

Pocket Quadrants

*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
PCM1770PWR	TSSOP	PW	16	2000	330.0	17.4	6.8	5.4	1.6	8.0	16.0	Q1
PCM1770RGAR	VQFN	RGA	20	2000	330.0	13.4	4.4	4.4	1.3	8.0	12.0	Q1
PCM1771PWR	TSSOP	PW	16	2000	330.0	17.4	6.8	5.4	1.6	8.0	16.0	Q1

TAPE AND REEL BOX DIMENSIONS

*All dimensions are nominal

Device	Package Type	Package Drawing	Pins	SPQ	Length (mm)	Width (mm)	Height (mm)
PCM1770PWR	TSSOP	PW	16	2000	367.0	367.0	38.0
PCM1770RGAR	VQFN	RGA	20	2000	367.0	367.0	35.0
PCM1771PWR	TSSOP	PW	16	2000	367.0	367.0	38.0

RGA (S–PQFP–N20)   PLASTIC QUAD FLATPACK
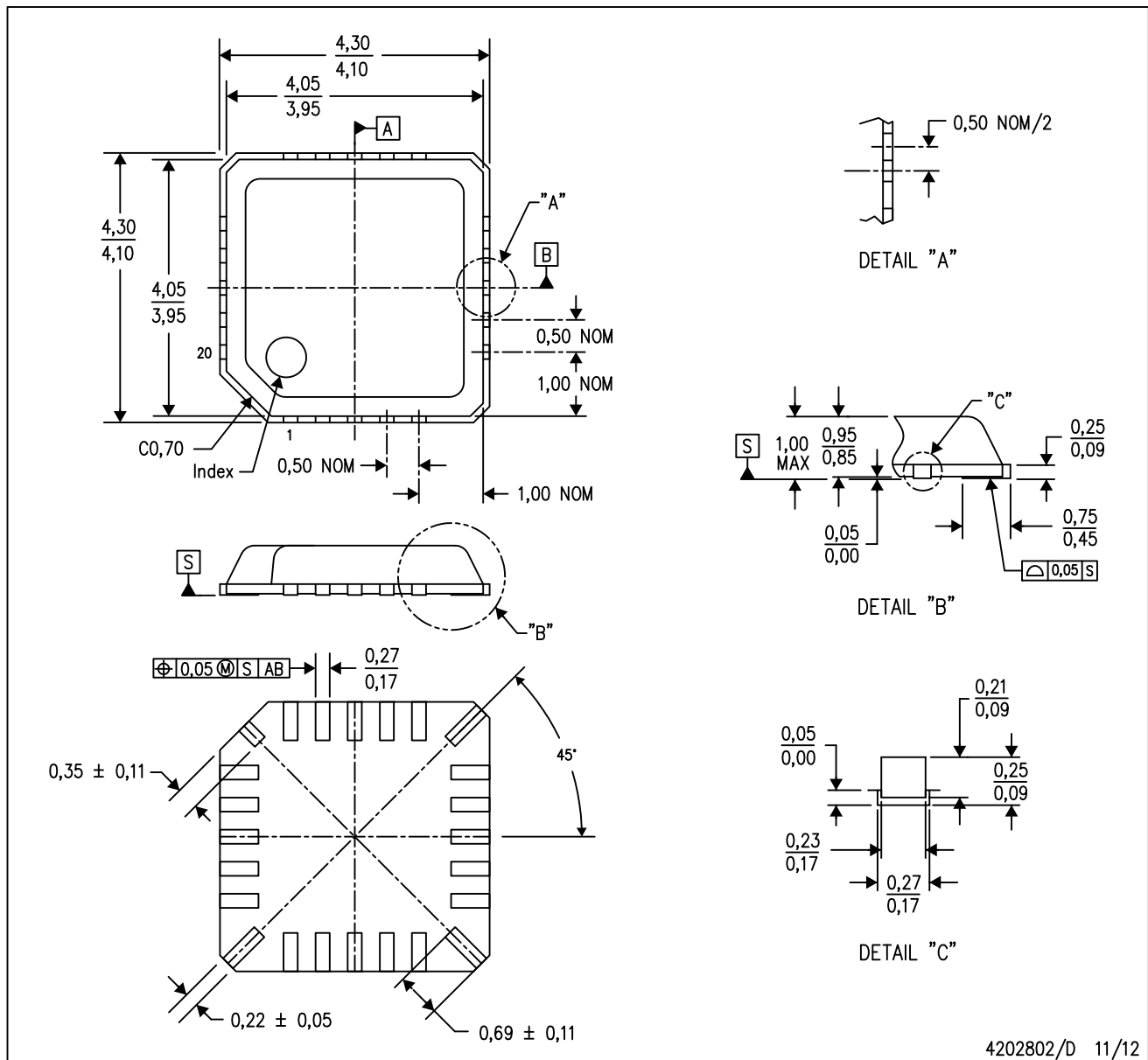
4,30
4,10

4,05
3,95

4,30
4,10

4,05
3,95

20

C0,70

Index

1

0,50 NOM

1,00 NOM

0,50 NOM

1,00 NOM

"A"

B

0,50 NOM/2

DETAIL "A"

S

"B"

⊕ 0,05 Ⓜ S AB

0,27
0,17

0,35 ± 0,11

45°

0,22 ± 0,05

0,69 ± 0,11

S 1,00 MAX

0,95
0,85

0,05
0,00

"C"

0,25
0,09

0,75
0,45

△ 0,05 S

DETAIL "B"

0,05
0,00

0,21
0,09

0,25
0,09

0,23
0,17

0,27
0,17

DETAIL "C"

4202802/D 11/12

NOTES: A. All linear dimensions are in millimeters.
         B. This drawing is subject to change without notice.
         C. These dimensions include package bend.
         D. Falls within EIAJ: EDR–7324.

PW (R−PDSO−G16)　　　　　　　PLASTIC SMALL OUTLINE

0,65

0,30
0,19

⊕ 0,10 Ⓜ

16　　　　　　9

D △ 4,50
4,30

6,60
6,20

1　　　　　　8

5,10
4,90

C △

0,15 NOM

Gage Plane

0°−8°

0,25

0,75
0,50

1,20 MAX

0,15
0,05

Seating Plane

⌒ 0,10

4040064−4/G　02/11

NOTES:　A.　All linear dimensions are in millimeters.　Dimensioning and tolerancing per ASME Y14.5M−1994.
　　　　　B.　This drawing is subject to change without notice.
　　　C △　Body length does not include mold flash, protrusions, or gate burrs.　Mold flash, protrusions, or gate burrs shall not exceed 0,15 each side.
　　　D △　Body width does not include interlead flash.　Interlead flash shall not exceed 0,25 each side.
　　　　　E.　Falls within JEDEC MO−153

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PW (R−PDSO−G16)   PLASTIC SMALL OUTLINE

Example Board Layout
(Note C)

←14x0,65→

5,60

Stencil Openings
(Note D)

←16x0,30→

←14x0,65→

16x1,55

5,60

Example
Non Soldermask Defined Pad

Example
Pad Geometry
(See Note C)

←0,35→

1,60

Example
Solder Mask Opening
(See Note E)

←0,07→
All Around

4211284−3/G   08/15

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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