



SLAS357 - DECEMBER 2001

PCM CODEC

FEATURES

- Designed for Analog and Digital Wireless Handsets, Voice-Enabled Terminals, and Telecommunications Applications
- 2.7-V to 3.3-V Operation
- Selectable 13-Bit Linear or 8-Bit μ-Law Companded Conversion
- Differential Microphone Input With External Gain Setting
- Differential Earphone Output Capable of Driving a 32- Ω to 8- Ω Load
- Programmable Volume Control in Linear Mode
- Microphone (MIC) and Earphone (EAR) Mute Functions
- Typical Power Dissipation of 0.03 mW in Power-Down Mode
- 2.048-MHz Master Clock Rate
- 300-Hz to 3.4-kHz Passband
- Low Profile 20-Terminal TSSOP Packaging

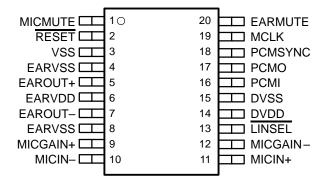
APPLICATIONS

- Digital Handset
- Digital Headset
- Cordless Phones
- Digital PABX
- Digital Voice Recording

DESCRIPTION

The TLV320AlC1106 PCM codec is designed to perform transmit encoding analog-to-digital (A/D) conversion, receive decoding digital-to-analog (D/A) conversion, and transmit and receive filtering for voice-band communications systems. The TLV320AlC1106 device operates in either the 13-bit linear or 8-bit companded (μ -law) mode. The PCM codec generates its own internal clocks from a 2.048-MHz master clock input.

PW PACKAGE (TOP VIEW)





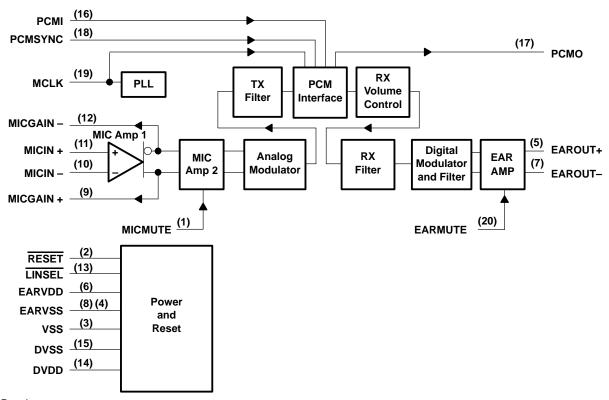
This device contains circuits to protect its inputs and outputs against damage due to high static voltages or electrostatic fields. These circuits have been qualified to protect this device against electrostatic discharges (ESD) of up to 2 kV according to MIL-STD-883C, Method 3015; however, it is advised that precautions be taken to avoid application of any voltage higher than maximum-rated voltages to these high-impedance circuits. During storage or handling, the device leads should be shorted together or the device should be placed in conductive foam. In a circuit, unused inputs should always be connected to an appropriated logic voltage level, preferably either VCC or ground. Specific guidelines for handling devices of this type are contained in the publication *Guidelines for Handling Electrostatic-Discharge-Sensitive (ESDS) Devices and Assemblies* available from Texas Instruments.



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.



functional block diagram



RX = Receive TX = Transmit

detailed description

power up/reset

An external reset must be applied to the active-low RESET terminal while MCLK is active to ensure reset at power up.

reference

A precision band-gap reference voltage is generated internally and supplies all required references to operate the transmit and receive channels.

phase-locked loop

The phase-locked loop generates the internal clock frequency required for internal digital filters and modulators by phase-locking to 2.048-MHz master clock input.

PCM interface

The PCM interface transmits and receives data at the PCMO and PCMI terminals, respectively. The data is transmitted or received at the MCLK speed once on each PCMSYN cycle. The PCMSYN can be driven by an external source that is derived from the master clock and used as an interrupt to the host controller.

microphone input

The microphone input circuit consists of two differential input/differential output amplifiers (MIC Amp 1 and MIC Amp 2). MIC Amp 1 is a low-noise differential amplifier capable of an externally set gain. MIC Amp 2 is a differential amplifier with a fixed gain of 6 dB.

analog modulator

The transmit channel modulator is a third-order sigma-delta design.

transmit filter

The transmit filter is a digital filter designed to meet Consultive Committee on International Telegraphy and Telephony (CCITT) G.714 requirements. The TLV320AlC1106 device operates in either the 13-bit linear or 8-bit companded μ -law mode.

receive filter

The receive (RX) filter is a digital filter that meets CCITT G.714 requirements. The TLV320AIC1106 device operates in either the 13-bit linear or 8-bit μ -law companded mode, which is selected at the LINSEL input.

receive volume control

In linear mode, the three least significant bits of the 16-bit PCMI data sample is used to control volume. The volume range is -18 dB to 3 dB in 3-dB steps.

digital modulator and filter

The second-order digital modulator and filter convert the received digital PCM data to the analog output required by the earphone interface.

earphone amplifiers

EAROUT is recommended for use as a differential output; however, it can be connected in single-ended topology as well. Clicks and pops are suppressed from the differential output.



Terminal Functions

TERMINAL				
TERMINAL	NO.	1/0	DESCRIPTION	
EARVSS	4	I	Analog ground for EAROUT+	
DVDD	14	I	Digital positive power supply	
DVSS	15	I	Digital negative power supply	
EARMUTE	20	I	Earphone mute	
EAROUT-	7	0	Earphone amplifier negative output	
EAROUT+	5	0	Earphone amplifier positive output	
EARVDD	6	I	Analog positive power supply for the earphone amplifiers	
EARVSS	8	I	alog ground for EAROUT-	
LINSEL	13	I	mpanding enable	
MCLK	19	I	Master system clock input (2.048 MHz) (digital)	
MICGAIN+	9	I	Microphone gain positive feedback	
MICGAIN-	12	I	Microphone gain negative feedback	
MICMUTE	1	I	Microphone mute	
MICIN-	10	I	Microphone negative input (–)	
MICIN+	11	I	Microphone positive input (+)	
PCMI	16	I	Receive PCM input	
РСМО	17	0	Transmit PCM output	
PCMSYNC	18	I	PCM frame synchronization	
RESET	2	I	Active-low reset	
VSS	3	ı	Ground return for band-gap internal reference	

absolute maximum ratings over operating free-air temperature range (unless otherwise noted)†

Supply voltage range, DVDD, EARVDD	0.5 V to 3.6 V
Output voltage range, V _O	0.5 V to 3.6 V
Input voltage range, V _I	0.5 V to 3.6 V
Continuous total power dissipation	See Dissipation Rating Table
Operating free air temperature range, T _A	–40°C to 85°C
Storage temperature range, T _{stq}	–65°C to 150°C
Lead temperature 1.6 mm from case for 10 seconds	

[†] 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.

DISSIPATION RATING TABLE

PACKAGE	T _A ≤ 25°C	DERATING FACTOR	T _A = 85°C
	POWER RATING	ABOVE T _A = 25°C	POWER RATING
PW	680 W	6.8 W/°C	270 W



recommended operating conditions (see Note 2)

	MIN	NOM	MAX	UNIT
Supply voltage, DVDD, EARVDD	2.7		3.3	V
High-level input voltage, V _{IH}	0.7xV _{DD}			V
Low-level input voltage, V _{IL}			$0.3 \times V_{DD}$	V
Load impedance between EAROUT+ and EAROUT-, RL		8 to 32		Ω
Input voltage, MICIN			0.9xV _{DD}	V
Operating free-air temperature, TA	-40	•	85	°C

- NOTES: 1. To avoid possible damage and resulting reliability problems to these CMOS devices, follow the *power-on initialization* paragraph, described in the *Principles of Operations*.
 - 2. Voltages are with respect to DVSS, and EARVSS.

electrical characteristics over recommended ranges of supply voltage and operating free-air temperature (unless otherwise noted)

supply current

	PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
I _{DD} Supply current from V _{DD}	Operating		5	7	mA	
	Power down, MCLK not present		10	30	μΑ	
t _{pu}	Power-up time from power down				10	ms

digital interface

PARAMETER		TEST CONDITIONS	MIN	TYP	MAX	UNIT
Vон	High-level output voltage, PCMO	$I_{OH} = -3.2 \text{ mA}, V_{DD} = 3.3 \text{ M}$	/ DVDD-0.25			V
V_{OL}	Low-level output voltage, PCMO	$I_{OL} = 3.2 \text{ mA}, \qquad V_{DD} = 3.3 \text{ mA}$	/		0.2	V
lн	High-level input current, any digital input	$V_I = 2.2 \text{ V to } V_{DD}$			10	μΑ
Ι _Ι L	Low-level input current, any digital input	V _I = 0 to 0.8 V			10	μΑ
Cl	Input capacitance				10	pF
Co	Output capacitance				20	pF

microphone interface

	PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
V _{IO}	Input offset voltage	See Note 3	-5		5	mV
I _{IB}	Input bias current		-250		250	nA
Ci	Input capacitance			5		pF
Vn	Microphone input referred noise, psophometric weighted	MIC Amp 1 gain = 23.5 dB, See Note 4		2.9	4	μV _{rms}
	MICMUTE		-80			dB

NOTES: 3. Measured while MICIN+ and MICIN- are connected together. Less than a 0.5-mV offset results in 0 value code on PCMOUT.

4. Configured as shown in Figure 3.



speaker interface

	PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Earphone AMP output power (see Note 5)	V_{DD} = 2.7 V, fully differential, 8-Ω load, 3-dBm0 output, volume control = -3 dB, PCMI data input to -4 dB level		161	200		
	V_{DD} = 2.7 V, fully differential, 16-Ω load, 3-dBm0 output, volume control = -3 dB, PCMI data input to -2 dB level		128	160	mW	
	Ediphone / will oduput power (add Note of	V_{DD} = 2.7 V, fully differential, 32- Ω load, 3-dBm0 output, volume control = -3 dB, PCMI data input to -1 dB level		81	100	
		3-dBm0 input, 8-Ω load		141	178	
Iomax	Maximum output current for EAROUT (rms)	3-dBm0 input, 16- Ω load		90	112	mA
		3-dBm0 input, $32-\Omega$ load		50	63	
	EARMUTE		-80			dB

NOTE 5: Maximum power is with a load impedance of -20%, at 25°C.

transmit gain and dynamic range, companded mode (µ-law) or linear mode selected (see Notes 6 and 7)

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Transmit reference-signal level (0dB)	Differential, MIC Amp 1 configured for 23.5 dB gain			88	mV_{pp}
Overload-signal level (3 dBm0)	Differential, MIC Amp 1 configured for 23.5 dB gain			124	mV _{pp}
Absolute gain error 0 dBm0 input signal, 2.7 V ≤ V _{DD} ≤ 3.3 V		-1		1	dB
	MICIN-, MICIN+ to PCMO at 3 dBm0 to -30 dBm0	-0.5		0.5	
Gain error with input level relative to gain at -10 dBm0 MICIN, MICIN+ to PCMO	MICIN-, MICIN+ to PCMO at -31 dBm0 to -45 dBm0	-1		1	dB
	MICIN-, MICIN+ to PCMO at -46 dBm0 to -55 dBm0	-1.2		1.2	

NOTES: 6. Unless otherwise noted, the analog input is 0 dB, 1020-Hz sine wave, where 0 dB is defined as the zero-reference point of the channel under test.

7. The reference signal level, which is input to the transmit channel, is defined as a value 3 dB below the full-scale value of 124-mV_{DD}.

transmit filter transfer, companded mode (µ-law) or linear mode selected

PARAMETER	TEST CONDITIONS	MIN	TYP MAX	UNIT
Gain relative to input signal gain at 1.02 kHz	f _{MIC} < 100 Hz	-0.5	0.5	
	f _{MIC} = 200 Hz	-0.5	0.5	
	f _{MIC} = 300 Hz to 3 kHz	-0.5	0.5	
	f _{MIC} = 3.4 kHz	-1.5	0	dB
	f _{MIC} = 4 kHz		-14	
	f _{MIC} = 4.6 kHz		-35	
	f _{MIC} = 8 kHz		-47	



transmit idle channel noise and distortion, companded mode (µ-law) selected

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Transmit idle channel noise, psophometrically weighted	MIC Amp 1 configured for 23.5-dB gain (see Note 8)		-80	-70	dBm0p
1	MICIN-, MICIN+ to PCMO at 3 dBm0	27			
	MICIN-, MICIN+ to PCMO at 0 dBm0	30			
	MICIN-, MICIN+ to PCMO at -5 dBm0	33			
Transmit signal-to-distortion ratio with 1.02-kHz sine-wave	MICIN-, MICIN+ to PCMO at -10 dBm0	36			ID 0
input	MICIN-, MICIN+ to PCMO at -20 dBm0	35			dBm0
	MICIN-, MICIN+ to PCMO at -30 dBm0	26			
	MICIN-, MICIN+ to PCMO at -40 dBm0	24			
	MICIN-, MICIN+ to PCMO at -45 dBm0	19			
Intermodulation distortion, 2-tone CCITT method, composite	CCITT G.712 (7.1), R2	49			10
power level, -13 dBm0	CCITT G.712 (7.2), R2	51		·	dB

NOTE 8: With recommended impedances and resistor tolerance of 1%

transmit idle channel noise and distortion, linear mode selected

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Transmit idle channel noise, psophometrically weighted	MIC Amp 1 configured for 23.5-dB gain (see Note 8)		-80	-74	dBm0 _p
	MICIN-, MICIN+ to PCMO at 3 dBm0	40	55		
	MICIN-, MICIN+ to PCMO at 0 dBm0	50	61		
	MICIN-, MICIN+ to PCMO at -5 dBm0	52	62		
Transmit signal-to-total distortion ratio with 1.02-kHz	MICIN-, MICIN+ to PCMO at -10 dBm0	56	66		.ID
sine-wave input	MICIN-, MICIN+ to PCMO at -20 dBm0	52	68		dB
	MICIN-, MICIN+ to PCMO at -30 dBm0	51	61		
	MICIN-, MICIN+ to PCMO at -40 dBm0	43	59		
	MICIN-, MICIN+ to PCMO at -45 dBm0	38	55		

NOTE 8: With recommended impedances and resistor tolerance of 1%

receive gain and dynamic range, linear or companded (µ-law) mode selected (see Note 9)

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
	Load = 8 Ω , volume control = -3 dB, PCMI data input to -4 dB level		3.2		
Overload-signal level (3 dB)	Load = 16 Ω , volume control = -3 dB, PCMI data input to -2 dB level		4.05		V_{pp}
	Load = 32 Ω , volume control = -3 dB, PCMI data input to -1 dB level		4.54		
Absolute gain error	0 dBm0 input signal, 2.7 V ≤ V _{DD} ≤ 3.3 V	-1		1	dB
	PCMI to EAROUT-, EAROUT+ at 3 dBm0 to -40 dBm0	-0.5		0.5	
Gain error with output level relative to gain at -10 dBm0	PCMI to EAROUT-, EAROUT+ at -41 dBm0 to -50 dBm0	-1		1	dB
Totalivo to gair at 10 abino	PCMI to EAROUT-, EAROUT+ at -51 dBm0 to -55 dBm0	-1.2		1.2	

NOTE 9: 1020-Hz input signal at PCMI, output measured differentially between EAROUT- and EAROUT+



receive filter transfer, companded mode (μ-law) or linear mode selected (MCLK = 2.048 MHz) (see Note 10)

PARAMETER	TEST CONDITIONS	MIN	TYP MA	UNIT
	fEAROUT < 100 Hz		-1	5
	fEAROUT = 200 Hz		_	5
	fEAROUT = 300 Hz to 3 kHz	-0.5	0.	5
Gain relative to input signal gain at 1.02-kHz	fEAROUT = 3.4 kHz	-1.5) dB
	fEAROUT = 4 kHz		-1	1
	fEAROUT = 4.6 kHz		-3	5
	fEAROUT = 8 kHz		-4	7

NOTE 10: Volume control = -3 dB, PCMI data input to -1 dB level (32- Ω load)

receive idle channel noise and distortion, companded mode (μ-law) selected (see Note 10)

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Receive noise, C-message weighted	PCMI = 11111111 (μ-law)		-90	-88	dBm0
	PCMI to EAROUT-, EAROUT+ at 3 dBm0	21			
	PCMI to EAROUT-, EAROUT+ at 0 dBm0	25			
	PCMI to EAROUT-, EAROUT+ at -5 dBm0	36			
Receive signal-to-distortion ratio with 1.02-kHz	PCMI to EAROUT-, EAROUT+ at -10 dBm0	43			
sine-wave input	PCMI to EAROUT-, EAROUT+ at -20 dBm0	40			dB
	PCMI to EAROUT-, EAROUT+ at -30 dBm0	38			
	PCMI to EAROUT-, EAROUT+ at -40 dBm0	28			
	PCMI to EAROUT-, EAROUT+ at -45 dBm0	23			

NOTE 10: Volume control = -3 dB, PCMI data input to -1 dB level (32- Ω load)

receive idle channel noise and distortion, linear mode selected (see Note 10)

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Receive noise, (20-Hz to 20-kHz brickwall window)	PCMI = 0000000000000		-83	-78	dBm0
	PCMI to EAROUT-, EAROUT+ at 3 dBm0	48	52		
	PCMI to EAROUT-, EAROUT+ at 0 dBm0	51	56		
Receive signal-to-distortion ratio with 1.02-kHz	PCMI to EAROUT-, EAROUT+ at -5 dBm0	57	59		
	PCMI to EAROUT-, EAROUT+ at -10 dBm0	55	62		-ID
sine-wave input (0-4 kHz)	PCMI to EAROUT-, EAROUT+ at -20 dBm0	51	53		dB
	PCMI to EAROUT-, EAROUT+ at -30 dBm0	45	47		
	PCMI to EAROUT-, EAROUT+ at -40 dBm0	42	47		
	PCMI to EAROUT-, EAROUT+ at -45 dBm0	35	45		
Intermodulation distortion, 2-tone CCITT method,	CCITT G.712 (7.1), R2	50			j
composite power level, -13 dBm0	CCITT G.712 (7.2), R2	54			dB

NOTE 10: Volume control = -3 dB, PCMI data input to -1 dB level (32- Ω load)



power supply rejection

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Supply voltage rejection, transmit channel	MICIN-, MICIN+ = 0 V, V_{DD} = 2.7 V + 100 mV _{pp} , f = 1 kHz, Resistor tolerance of 1%		-74	-50	dB
Supply voltage rejection, receive channel (differential)	PCM code = positive zero, V_{DD} = 2.7 V + 100 mV _{pp} , f = 1 kHz, Resistor tolerance of 1%		-80	-65	dB

crosstalk attenuation, linear mode selected

PARAMETER	TEST CONDITIONS	MIN	TYP	MAX	UNIT
Crosstalk attenuation, transmit-to-receive (differential)	MICIN-, MICIN+ = 0 dB, f = 300 Hz to 3400 Hz measured differentially between EAROUT- and EAROUT+	70			dB
Crosstalk attenuation, receive-to-transmit	PCMI = 0 dBm0, f = 300 Hz to 3400 Hz measured at PCMO	70	•		dB

timing requirements

clock

		MIN	NOM	MAX	UNIT
t _t	Transition time, MCLK			10	ns
fmclk	MCLK frequency		2.048		MHz
	MCLK jitter			37%	
	MCLK clock cycles per PCMSYN frame	256		256	cycles

transmit (see Figure 1)

		MIN	MAX	UNIT
tsu(PCMSYN)	Setup time, PCMSYN high before MCLK \downarrow	20	t _C (MCLK)-20	ns
th(PCMSYN)	Hold time, PCMSYN high after MCLK \downarrow	20	t _{c(MCLK)} -20	

receive (see Figure 2)

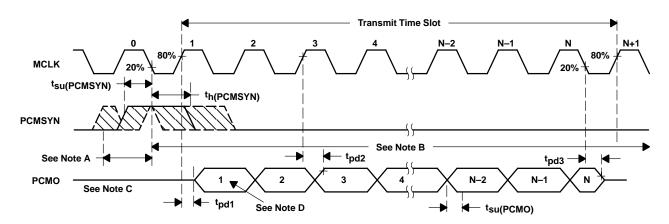
		MIN	MAX	UNIT
t _{su} (PCSYN)	Setup time, PCMSYN high before MCLK \downarrow	20	t _{c(MCLK)} -20	ns
th(PCSYN)	Hold time, PCMSYN high after MCLK \downarrow	20	t _c (MCLK)-20	ns
t _{su} (PCMI)	Setup time, PCMI high or low before MCLK \downarrow	20		ns
th(PCMI)	Hold time, PCMI high or low after MCLK ↓	20		ns

switching characteristics over recommended operating conditions, C_L max = 10 pF (see Figure 1)

		MIN	MAX	UNIT
tpd1	Propagation delay time, MCLK bit 1 high to PCMO bit 1 valid		35	ns
t _{pd2}	Propagation delay time, MCLK high to PCMO valid, bits 2 to n		35	ns
t _{pd3}	Propagation delay time, MCLK bit n low to PCMO bit n Hi-Z	30		ns



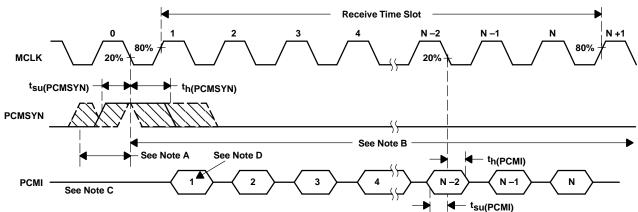
PARAMETER MEASUREMENT INFORMATION



NOTES: A. This window is allowed for PCMSYN high.

- B. This window is allowed for PCMSYN low (th(PCMSYN) max determined by data collision considerations).
- C. Transitions are measured at 50%.
- D. Bit 1 = Most significant bit (MSB), Bit N = Least significant bit (LSB)

Figure 1. Transmit Timing Diagram



NOTES: A. This window is allowed for PCMSYN high.

- B. This window is allowed for PCMSYN low.
- C. Transitions are measured at 50%.
- D. Bit 1 = Most significant bit (MSB), Bit N = Least significant bit (LSB)

Figure 2. Receive Timing Diagram



PRINCIPLES OF OPERATION

power-up initialization

An external reset with a minimum pulse width of 500 ns must be applied to the active-low RESET terminal with MCLK active to ensure reset upon power up.

Table 1. Power-Up and Power-Down Power Consumption (V_{DD} = 2.7 V, Earphone Amplifier Loaded)

DEVICE STATUS	MAXIMUM POWER CONSUMPTION
Power up	16.2 mW
Power down	81 μW

The loss of MCLK (no transition detected) automatically enters the device into a power-down state with PCMO in the high-impedance state. If an asynchronous power down occurs during a pulse code modulation (PCM) data transmit cycle, the PCM interface remains powered up until the PCM data is completely transferred.

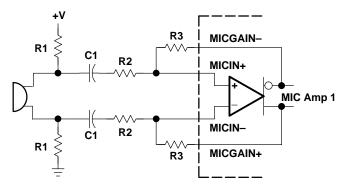
conversion laws

The device can be programmed either for a 13-bit linear or 8-bit (μ -law) companding mode. The companding operation approximates the CCITT G.711 recommendation. The linear mode operation uses a 13-bit twos complement format. Linear mode is selected with $\overline{\text{LINSEL}}$ low. $\overline{\text{LINSEL}}$ is high for companding.

transmit operation

microphone input

The microphone input stage is a low-noise differential amplifier. The microphone must be capacitively coupled to the MICIN– and MICIN+ terminals. Preamplifier (MIC Amp 1) gain is determined by selection of external resistors R2 and R3. To achieve the recommended gain setting of 23.5 dB for MIC Amp 1, resistor values of R2 = 34 k Ω and R3 = 510 k Ω are suggested. A 1% tolerance is recommended for all resistors to meet the specification. The recommended range for R2 is 34–100 k Ω . For values above 100 k Ω , the noise performance of the channel is degraded.



R1 = $2 k\Omega$ C1 = $0.22 \mu F$

MIC Amp 1 Gain in dB = 20 $\log \left(\frac{R3}{R2}\right)$

Figure 3. Typical Microphone Interface



PRINCIPLES OF OPERATION

microphone mute function

Transmit channel muting can be selected by setting MICMUTE high. Muting provides 80-dB attenuation of the input microphone signal.

receive operation

earphone amplifier

The analog signal is routed to the earphone amplifier differential output (EAROUT– or EAROUT+), which is capable of driving a load as low as 8Ω . EAROUT is recommended for use as a differential output.

earphone mute function

Receive channel muting can be selected by setting the EARMUTE terminal to high.

receive PCM data format

Companded mode: 8 bits are received, the MSB first

Linear mode: 13 bits are received, the MSB first

Table 2. Receive Data Bit Definitions

BIT NO.	COMPANDED MODE	LINEAR MODE
1	CD7	LD12
2	CD6	LD11
3	CD5	LD10
4	CD4	LD9
5	CD3	LD8
6	CD2	LD7
7	CD1	LD6
8	CD0	LD5
9	_	LD4
10	_	LD3
11	_	LD2
12	_	LD1
13	_	LD0
14	_	RXVOL2
15	_	RXVOL1
16	_	RXVOL0



PRINCIPLES OF OPERATION

receive volume control

In linear mode, RXVOL [2:0] PCM data bits are used for volume control according to Table 3. Volume control bits must be sent on PCMI for each 13-bit receive word. In companded mode, volume control is fixed at 0 dB.

Table 3. Volume Control Bit Definition in Linear Mode

RXVOL [2:0]	GAIN SETTING
000	3 dB
001	0 dB
010	−3 dB
011	−6 dB
100	−9 dB
101	-12 dB
110	-15 dB
111	-18 dB

support section

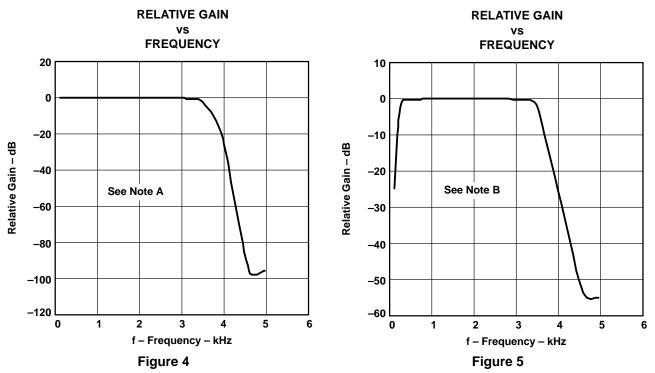
The clock generator and control circuit uses the master clock input (MCLK) to generate internal clocks to drive internal counters, filters, and converters.

clock frequencies and sample rates

A fixed PCMSYN rate of 8 kHz determines the sampling rate. The PCMSYN signal must be derived from the master clock. The divide ratio must be set to 256 for the device to work properly.

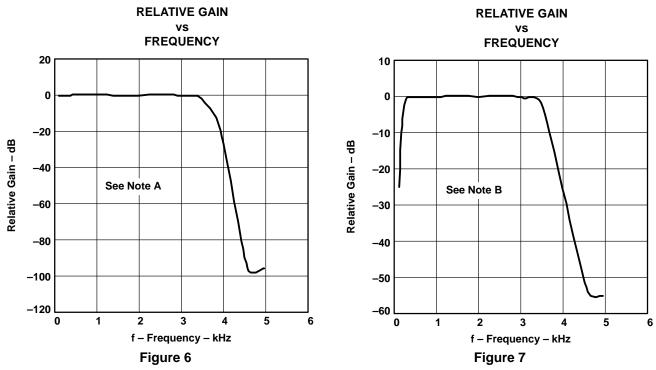


TYPICAL CHARACTERISTICS



NOTES: A. Transmit channel frequency response shown relative to the gain at a 1.02-kHz input signal in linear mode.

B. Receive channel frequency response shown relative to the gain at a 1.02-kHz input signal in linear mode.

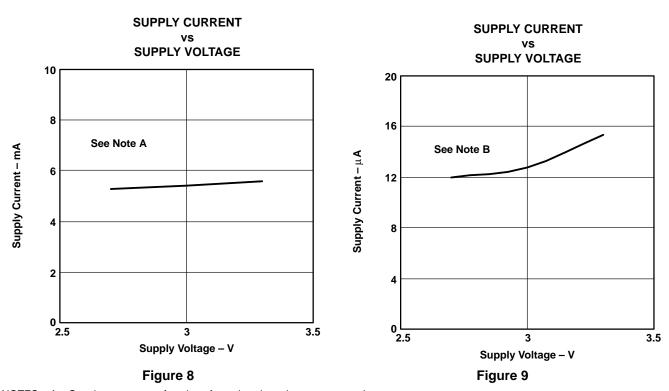


NOTES: A. Transmit channel frequency response shown relative to the gain at a 1.02-kHz input signal in μ -Law mode.

B. Receive channel frequency response shown relative to the gain at a 1.02-kHz input signal in μ -Law mode.



TYPICAL CHARACTERISTICS



NOTES: A. Supply current as a function of supply voltage in power-up mode.

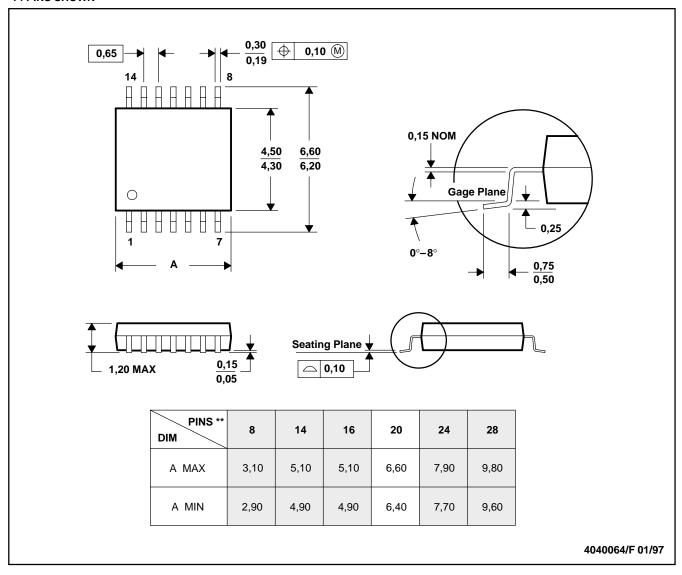
B. Supply current as a function of supply voltage in power-down mode.

MECHANICAL DATA

PW (R-PDSO-G**)

14 PINS SHOWN

PLASTIC SMALL-OUTLINE PACKAGE



NOTES: A. All linear dimensions are in millimeters.

B. This drawing is subject to change without notice.

C. Body dimensions do not include mold flash or protrusion not to exceed 0,15.

D. Falls within JEDEC MO-153



IMPORTANT NOTICE

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