

### LM4930 Boomer® Audio Power Amplifier Series

## Audio Subsystem with Stereo Headphone & Mono Speaker Amplifiers

### **General Description**

The LM4930 is an integrated audio subsystem that supports voice and digital audio functions. The LM4930 includes a high quality I<sup>2</sup>S input stereo DAC, a voice band codec, a stereo headphone amplifier and a high-power mono speaker amplifier. It is primarily designed for demanding applications in mobile phones and other portable devices.

The LM4930 features an I²S serial interface for full range audio, a 16-bit PCM bi-directional serial interface for the voice band codec and an two-wire interface for control. The full range music path features an SNR of 86dB with a 16-bit 48kHz input. The stereo DAC can also be used while the voice codec is in use. The headphone amplifier delivers  $25\text{mW}_{\text{RMS}}$  to a  $32\Omega$  single-ended stereo load with less than 0.5% distortion (THD+N) when AV\_DD = 3V. The mono speaker amplifier delivers up to 330mW into an  $8\Omega$  load with less than 1% distortion when AV\_DD = 3V.

The LM4930 employs advanced techniques to reduce power consumption, to reduce controller overhead and to eliminate click and pop. Boomer audio power amplifiers were designed specifically to provide high quality output power with a minimal amount of external components. It is, therefore, ideally suited for mobile phone and other low voltage applications where minimal power consumption is a primary requirement.

### **Key Specifications**

■  $P_{LS OUT}$  at  $AV_{DD} = 5.0V$ ,  $8\Omega$ 

1% THD+N	1W (typ)
$P_{LS~OUT}$ at $AV_{DD} = 3.0V$ , $8\Omega$	
1% THD+N	330mW (typ)
$P_{H/P \text{ OUT}}$ at $AV_{DD} = 3.0V$ , $32\Omega$	
0.5% THD+N	25mW (typ)
Supply voltage range	
DV <sub>DD</sub> (Note 8)	2.6V to 4.5V

D V DD (Note o)	2.01 to 1.01
AV <sub>DD</sub> (Note 8)	2.6V to 5.5V
■ Total shutdown current	2μA (typ)
■ PSRR at 217Hz, AV <sub>DD</sub> = 3V	50dB (typ)

### **Features**

- 16-bit resolution 48kHz stereo DAC
- 16-bit resolution 8kHz voice codec
- I<sup>2</sup>S digital audio data serial interface
- Two-wire serial control interface
- PCM voice audio data serial interface
- 25mW/channel stereo headphone amplifier
- 330mW mono 8Ω amplifier (at AV<sub>DD</sub> = 3.0V)
- 32-step volume control for audio output amplifiers
- No snubber networks or bootstrap capacitors are required by the headphone or hands-free amplifiers
- Digital sidetone generation with adjustable attenuation
- Gain controllable headphone amp, mono BTL amp, mic preamp
- Available in the 36-bump micro SMD and 44-lead LLP packages

### **Applications**

- Mobile Phones
- Mobile/low power audio appliances
- PDAs

## Typical Application **T**jpical Application

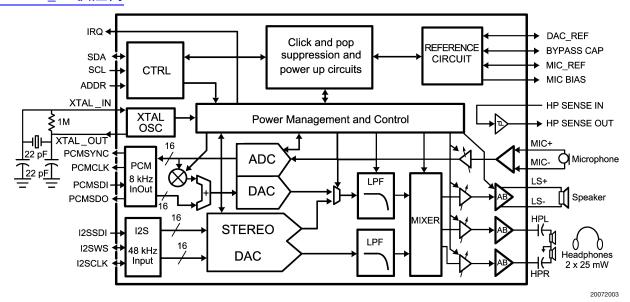
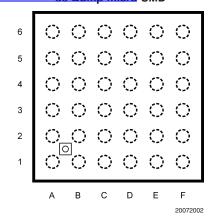


FIGURE 1. Typical I<sup>2</sup>S + Voice codec application circuit for mobile phones

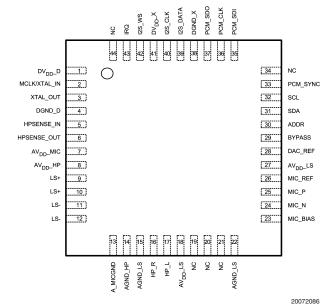
### **Connection Diagrams**

查询"LM4930<u>604"</u>供应商 SMD



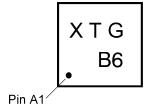
Top View Order Number LM4930ITL See NS Package Number MKT - TLA36KRA

44 - Lead LLP



Top View
Order Number LM4930LQ
See NS Package Number MKT - LQA44A

micro SMD Marking



20072056

Top View
X - Date Code
T - Die Traceability
G - Boomer Family
B6 - LM4930ITL

44 - Lead LLP Marking



2007208

Top View
N - National Logo
U - Wafer Fab Code
Z - Assembly Plant Code
XY - Two Digit Date Code
TT - Die Run Code
L4930LQ - LM4930LQ

### Pin Descriptions

'Ļ <mark>М4</mark> 9	30,4,04,"供应商	Microphone positive differential input
A2	MIC_N	Microphone negative differential input
А3	AVDD_MIC	Analog V <sub>dd</sub> for microphone preamp
A4	DAC_REF	D/A converter reference voltage
A5	SDA	Two-wire control interface serial data pin
A6	SCL	Two-wire control interface serial clock pin
B1	AGND_MIC	Analog ground for microphone preamp
B2	MIC_BIAS	Microphone bias supply output (2V)
В3	MIC_REF	Internal fixed-reference bypass capacitor decoupling pin
B4	ADDR	Control bus address select pin
B5	PCM_SDI	PCM serial data in
B6	PCM_CLK	PCM Serial clock pin
C1	AVDD_HP	Analog V <sub>dd</sub> for headphone amplifier
C2	NC	No Connect
СЗ	BYPASS	Half-supply bypass capacitor decoupling pin
C4	PCM_SYNC	PCM Frame sync pin
C5	I2S_DATA	I <sup>2</sup> S serial data input
C6	DGND_D	Digital ground
D1	HP_L	Headphone amplifier connection (Left)
D2	HP_R	Headphone amplifier connection (Right)
D3	HPSENSE_IN	Connection for sense pin of headphone jack
D4	PCM_SDO	PCM serial data out
D5	I2S_CLK	I <sup>2</sup> S serial bit clock
D6	DVDD_D	Digital V <sub>dd</sub>
E1	AGND_HP	Analog ground for headphone amplifier
E2	LS-	Loudspeaker amplifier BTL negative out (-)
E3	HPSENSE_OUT	Logic output pin to indicate headphone connection status. Outputs logic high when
		HPSENSE_IN is high and outputs logic low when HPSENSE_IN is low. See Figure 5 for suggested application circuit
E4	IRQ	LM4930 mode status indicator pin
E5	I2S_WS	I <sup>2</sup> S word select
E6	XTAL_OUT	Negative feedback source for external crystal MCLK
F1	AGND_LS	Analog ground for loudspeaker amplifier
F2	LS+	Loudspeaker amplifier BTL positive out (+)
F3	AVDD_LS	Analog V <sub>DD</sub> for loudspeaker amplifier
F4	DGND_X	Digital ground
F5	DVDD_X	Digital V <sub>DD</sub>
F6	MCLK/XTAL_IN	12.288MHz or 24.576MHz Master Clock from crystal (via XTAL OUT) or external source
_ ' '	WOLIVATAL_IIV	12.200mi 2 01 27.07 0mi 12 master clock from crystal (via ATAL COT) of external source

This interface is used to configure the operating mode, digital interfaces, and delta-sigma modulators. The LM4930 is

controlled by writing information into a series of write-only registers, each with its own unique 7 bit address. The following registers are programmable:

#### **BASIC CONFIG REGISTER**

This register is used to configure the I2S and PCM interfaces as well as the 48kHz DAC module. The 7 bit address for the BASICCONFIG register is XX10000. (X = 0 if ADDR is set to logic 0)(X = 1 if ADDR is set to logic 1)

### BASIC CONFIGURATION (XX1000). (Set = logic 1, Clear = logic 0)

BIT	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0
RESET	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0

Address	Register	Description	on									
3:0	MODE	new mode	30 can be placed in on is selected the LM493 nagement profile autom	0 will change operat	tion silently and will re	-configure the						
		Mode	Mono Speaker Amplifier Source	Headphone Left Source	Headphone Right Source	Comment						
		0000	None	None	None	Powerdown mode						
		0001	None	None	None	Standby mode						
		0010	Voice	None	None	Mono speaker mode						
		0011	None	Voice	Voice	Headphone call mode						
		0100	Voice	Voice	Voice	Conference call mode						
		0101	Audio (L+R)	None	None	L+R mixed to mono speaker						
		0110	None	Audio (Left)	Audio (Right)	Headphone stereo audio						
		0111	Audio (L+R)	Audio (Left)	Audio (Right)	L+R mixed to mono speaker + stereo headphone audio						
		1000	Audio (Left)	Voice	Voice	Mixed Mode						
		1001	Voice + Audio (Left)	Voice	Voice	Mixed mode						
		1010	Voice	Audio (Left)	Audio (Left)	Mixed Mode						
4	SOFT_RESET	Resets the	e LM4930, excluding th	e control registers								
5	PCM_LONG	If set the F	PCM interface uses a lo	ong frame sync. (No	te 12)							
6	PCM_COMPANDED	If set the 8	B MSBs are presumed	to be companded da	ata and the 8 LSBs ar	e ignored. (Note						
7	PCM_LAW	If set, the	companded G711 data	is set to be A-law, e	else µ-law is assumed	(Note 12)						
8:9	PCM_SYNC_MODE		h), 2 (01h) or 4(10h) 1 rames. (Note 12)	6 bit frames per syn	c. The PCM_SDO pin	is tri-stated during						
10	PCM_ALWAYS_ON		ould be set if another of slock and sync signals	=								
11	I2S_M/S	I2S maste	r or slave select. If set	then I2S = master. (	Cleared = slave							
12	I2S_RES	I2S resolu	tion select. If set then 3	32 bits per frame. If	cleared then 16 bits p	er frame						
13	RSVD	RESERVED (Note 13)										
14	RSVD	RESERVED (Note 13)										
15	RSVD	RESERVED (Note 13)										

## 

### **VOICE/TEST CONFIG REGISTERS**

This register configures the voiceband codec, sidetone attenuation, and selected control functions. The 7 bit address for the VOICE TESTCONFIG register is XX10001. (X = 0 if ADDR is set to logic 0) (X = 1 if ADDR is set to logic 1)

### VOICETESTCONFIG (XX10001). (Set = logic 1, Clear = logic 0)

BIT	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	
RESET	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	
			•	•	•	•			•			•	•	•	and turns the in turns the interest in		
Address	Register		Desci	ription													
0	CLASS			_						nal clas	s D or	linear	amplifie	er and t	turns t	he	
			BTL s	BTL speaker output into a buffer. (Note 12)										er and turns the ws: ion  cations where des when voice is sidetone is preser 88MHz crystal.  Ite mode changes  or powering down muted			
4:1	SIDESTON	NE_ATTEN	Progra	ams the	e atten	uation	of the o	digital s	idetone	. Atten	uation	is set a	as follov	vs:			
			4:1			Sidetone 4:1 5			Sidet	one At	near amplifier and turns the set as follows:  ne Attenuation  ds-free applications where muted in modes when voice therwise the sidetone is prestig is for 12.288MHz crystal.  tee immediate mode change orrect turn-off delay and some and already muted						
			0000		Mute			1000			-9dB		set as follows:  ne Attenuation  Is-free applications where muted in modes when voice herwise the sidetone is preside is for 12.288MHz crystal.  tee immediate mode change				
			0001		-30dB	,		1001			-6dB		ear amplifier and turns the set as follows:  e Attenuation  -free applications where uted in modes when voice is erwise the sidetone is present is for 12.288MHz crystal.  ee immediate mode changes rect turn-off delay and				
			0010		-27dB	,		1010			-3dB	linear amplifier and turns the is set as follows:  one Attenuation					
			0011		-24dB	}		1011			0dB						
			0100		-21dB	,		1100			Mute						
			0101		-18dB	,		1101			Mute						
			0110		-15dB	,		1110			Mute						
			0111		-12dB	1		1111			Mute						
5	AUTOSIDI	Ε	This feature is included for use with the mono speaker in hands-free applications w														
					•							ways muted in modes when voice					
			1 -			-						otherw	ise the	sidetor	ne is p	resent	
									n conro								
6	CLOCK_D	ilV	If set, (Note		for the	use of	f a 24.5	576MH:	z crysta	l. Defa	ult setti	ing is f	or 12.2	88MHz	crysta	al.	
7	ZXD_DISA	ABLE	Disab	les the	zero c	rossing	detect	in the	stereo	DAC to	guara	ntee ir	nmedia	te mod	e chai	nges	
			rather	than v	aiting	for a ze	ero cro	ss. (No	te 11)								
8:9	RSVD		RESE	RVED	(Note	13)											
10:11	CAP_SIZE		1					-			_	correct	turn-of	f delay	and		
			<u> </u>	op per			ue is s		ollows:	`							
			10:11		Delay	'		Вура	ss Cap	acitor	Size						
			00		25ms			0.1µF									
			01		50ms			0.39µ	F								
			10		85ms			1µF									
			11		RESE	RVED		RESE	RVED								
12	ZXDS_SL0															wn	
13	MUTE_LS																
14	MUTE_HP	•	If set,	mutes	the he	adphor	ne amp	lifier in	any mo	ode wh	ere it is	not a	Iready r	nuted			

If set, mutes the microphone preamp

www.national.com

15

MUTE\_MIC

### System Control Registers (Continued) 查询"LM4930\_04"供应商

### **GAIN CONFIG REGISTERS**

This register is used to control the gain of the headphone amplifier, the loudspeaker amplifier, and the microphone preamplifier. The 7 bit address for the GAINCONFIG register is XX10010. (X = 0 if ADDR is set to logic 0) (X = 1 if ADDR is set

### GAINCONFIG (XX10010). (Set = logic 1, Clear = logic 0)

BIT	15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0
RESET	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0	0

NESEI	10	10	0	0	10	10	0	10	10	10	0	0	10	0	0	10	
Address	Register		Desc	ription													
4:0	LOUDSPK	R_GAIN				of the I	oudspe	aker a	mplifier	. Gain i	is set a	as follo	follows: hone Gain 3				
4:0 L			4:0		Loud	Ispeake	er	4:0	4:0			Loudspeaker Gain					
			00000	)	-34.5	dB		1000	0		-10.5	dB					
			0000	1	-33dE	3		1000	1		-9dB						
			00010	)	-31.5	dB		1001	0		-7.5d	В					
			00011	1	-30dE	3		1001	1		-6dB						
			00100	)	-28.5	dB		1010	0		-4.5d	В					
			0010	1	-27dE	3		1010	1		-3dB						
			00110	)	-25.5	dB		10110	)		-1.5d	В					
			00111		-24dE	3		10111	1		0dB						
			01000	)	-22.5	dB		11000	)		1.5dE	3					
			0100	1	-21dE	3		1100	1		3dB						
			01010	)	-19.5	dB		11010	)		4.5dE	3					
			01011	1	-18dE	3		11011	1		6dB						
			01100	)	-16.5	dB		11100	)		7.5dE	3		er Gain  SS:  Gain			
			01101	1	-15dE	3		11101	1		9dB						
			01110	)	-13.5	dB		11110	)		10.50	lB					
			01111		-12dE	3		11111			12dB						
9:5	HP_GAIN			ams th		of the h		one an	nplifier.	Gain is	set as	s follov	NS:				
			9:5		Headphone Gain			9:5			Head	phone	e Gain				
			00000	)	-46dE	3		1000	0		-22.5	dB					
			0000	1	-45dE	3		1000	1		-21dE	3					
			00010	)	-43.5	dB		1001	0		-19.5	dB					
			00011	1	-42db	)		1001	1		-18dE	3					
			00100	)	-40.5	dB		1010	0		-16.5	dB					
			0010	1	-39dE			1010	1		-15dE						
			00110	)	-37.5	dB		10110	)		-13.5	dB					
			00111		-36dE	3		10111	1		-12dE	3					
			01000	)	-34.5	dB		11000	)		-10.5	dB					
			0100	1	-33dE	3		1100	1		-9dB						
			01010	)	-31.5	dB		11010	)		-7.5d	В					
			01011		-30dE			11011			-6dB						
			01100		-28.5	dB		11100			-4.5dB						
			01101	1	-27dE	3		11101	1		-3dB						
			01110		-25.5	dB		11110	)		-1.5d	В					
			01111		-24dE	3		11111			0dB						
13:10	MIC_GAIN	1	Progr	ams th	e gain	of the r	nicroph	none ar	mplifier.	Gain is	s set a	s follo	ws:				

# System Control Registers (Continued) jp "LM4930\_04"供应商 GAIN CONFIG

### GAIN CONFIG REGISTERS (Continued)

		13:10	Mic Preamp Gain
		0000	17dB
		0001	19dB
		0010	21dB
		0011	23dB
		0100	25dB
		0101	27dB
		0110	29dB
		0111	31dB
		1000	33dB
		1001	35dB
		1010	37dB
		1011	39dB
		1100	41dB
		1101	43dB
		1110	45dB
		1111	47dB
15:14	RSVD	RESERVED (Note 1	3)

Timing Diagrams 查询"LM4930\_04"供应商

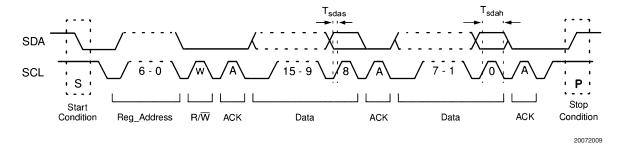
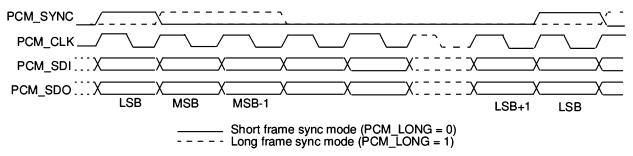
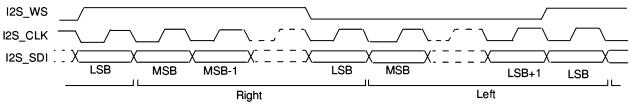


FIGURE 2. Two-wire control Interface Timing Diagram



20072008

FIGURE 3. PCM Receive Timing Diagram



20072007

FIGURE 4. I<sup>2</sup>S Transmit Timing Diagram

9

Absolute Maximum Ratings (Notes 1, 2) please contact the National Semiconductor Sales Office/ Distributors for availability and specifications.

> Analog Supply Voltage 6.0V Digital Storage Supply Voltage 6.0V Storage temperature -65°C to +150°C Power Dissipation (Note 3) Internally Limited **ESD Susceptibility**

Human Body Model (Note 4) 2000V 200V Machine Model (Note 5) Junction temperature 150°C

Thermal Resistance

 $\theta_{\mathsf{JA}}$  - TLA36KRA 105°C/W  $\theta_{JA}$  - LQA44A (Note 17) 27°C/W

### **Operating Ratings** (Note 3)

Temperature Range

 $-30^{\circ}C \leq T_{A} \leq +85^{\circ}C$  $T_{MIN} \leq T_A \leq T_{MAX}$ 

Supply Voltage

2.6V - 4.5V  $\mathrm{DV}_{\mathrm{DD}}$  (Note 8)

2.6V - 5.5V AV<sub>DD</sub> (Note 8)

### Electrical Characteristics DV<sub>DD</sub> = 3.3V, AV<sub>DD</sub> = 5V, R<sub>LHP</sub> = 32 $\Omega$ , R<sub>LHF</sub> = 8 $\Omega$

(Notes 1, 2, 8)

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for T<sub>A</sub>= 25°C.

Symbol	Parameter	Conditions	LM4	1930	Units	
			Typical (Note 6)	Limits (Notes 7, 15)	(Limits)	
		f <sub>MCLK</sub> = 12.288MHz				
		Output Mode = "0010" Output Mode = "0011" Output Mode = "0100"	2			
DI <sub>DD</sub>	Digital Power Supply Current	Output Mode = "0101" Output Mode = "0110" Output Mode = "0111"	4.4			
		Output Mode = "1000" Output Mode = "1001" Output Mode = "1010"	4.9	8	mA (max)	
		f <sub>MCLK</sub> = 12.288MHz; No Load				
		Output Mode = "0010"	7.0			
		Output Mode = "0011"	6.3			
		Output Mode = "0100"	8.0			
A.I.	Analog Power Supply Quiescent	Output Mode = "0101"	8.2			
$AI_{DD}$	Current	Output Mode = "0110"	7.4			
		Output Mode = "0111"	8.7			
		Output Mode = "1000" Output Mode = "1001" Output Mode = "1010"	9.5	14	mA (max)	
DI <sub>SD</sub>	Digital Powerdown Current	f <sub>MCLK</sub> = 12.288MHz Output Mode = "0000" Powerdown Mode	1	7	μA (max)	
Al <sub>SD</sub>	Analog Powerdown Current	f <sub>MCLK</sub> = 12.288MHz Output Mode = "0000" Powerdown Mode	1	2	μA (max)	
DI <sub>ST</sub>	Digital Standby Current	f <sub>MCLK</sub> = 12.288MHz Output Mode = "0001" Standby Mode	1.4	2	mA (max)	
Al <sub>ST</sub>	Analog Standby Current	f <sub>MCLK</sub> = 12.288MHz Output Mode = "0001" Standby Mode	230	1000	μA (max)	
$V_{FS\_LS}$	Full-Scale Output Voltage (Mono speaker amplifier)	CLASS = 0; 0dB gain setting; $8\Omega$ BTL load (Note 10)	2.5		$V_{P-P}$	
$V_{FS\_HP}$	Full-Scale Output Voltage (Headphone amplifier)	OdB gain setting; $32\Omega$ Stereo Load (Note 10)	2.5		$V_{P-P}$	

## Electrical Characteristics DV<sub>DD</sub> = 3.3V, AV<sub>DD</sub> = 5V, R<sub>LHP</sub> = 32 $\Omega$ , R<sub>LHF</sub> = 8 $\Omega$ (No語句上的4880ப்பெர்

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for  $T_A = 25$ °C.

Symbol	Parameter	Conditions	LM4	1930	Units (Limits)	
			Typical (Note 6)	Limits (Notes 7, 15)		
V <sub>MIC_BIAS</sub>	Mic Bias Voltage		2.0		V	
THD+N	Headphone Amplifier Total Harmonic Motion Distortion + Noise	$f_{\text{IN}}$ = 1 kHz, $P_{\text{OUT}}$ = 7.5mW; 32 $\Omega$ Stereo Load	0.07		%	
ОНР	Headphone Amplifier Output Power	THD+N = $0.5\%$ , $f_{OUT} = 1kHz$	27	20	mW (min)	
P <sub>OLS</sub>	Mono Speaker Amplifier Output Power	THD+N = 1%, $f_{OUT} = 1kHz$	1		W	
PSRR	Power Supply Rejection Ratio	$C_{BYPASS} = 1.0 \mu F$ $C_{DAC\_REF} = 1.0 \mu F$ $V_{RIPPLE} = 200 m V_{P-P} @ 217 Hz, MIC_P,$ $MIC\_N$ terminated with $10\Omega$ to ground	55	45	dB (min)	
SNR (Voice)	Signal-to-Noise Ratio (Voice DAC Path)	Signal = Vo at f = 1kHz @1% THD+N, 32Ω Stereo Load; Noise = digital zero, A-weighted, 0dB gain setting	72		dB	
SNR (Music)	Signal-to-Noise Ratio (Music Audio Path)	Signal = Vo at f = 1kHz @1% THD+N, 32Ω Stereo Load; Noise = digital zero, A-weighted; 0dB gain setting	86		dB	
DR (Voice)	Dynamic Range (Voice DAC Path)	Signal = Vo at f = 1kHz @1% THD+N, $32\Omega$ Stereo Load; Noise for -60dBFS digital input; A-weighted; 0dB gain setting	72		dB	
DR (Music)	Dynamic Range (Music Audio Path)	Signal = Vo at f=1kHz @1% THD+N, 32Ω Stereo Load; Noise for -60dBFS digital input; A-weighted, 0dB gain setting	86		dB	
SNR <sub>ADC</sub>	Signal-to-Noise Ratio (Voice ADC Path)	Reference signal = 0dBFS MIC_P, MIC_N terminated with 10Ω to ground; A-weighted; 47dB MIC preamp gain setting	75		dΒ	
OR <sub>ADC</sub>	Dynamic Range (Voice ADC Path)	Reference signal = 0dBFS Noise for -60dBFS digital input; A-weighted; 47dB MIC preamp gain setting	75		dΒ	
X <sub>TALK</sub>	Stereo Channel-to-Channel Crosstalk	$f_S = 48kHz$ , $f_{IN} = 1kHz$ sinewave at $-3dB_{FS}$	75		dB	
V <sub>MIC-IN</sub>	Maximum Differential MIC Input Voltage	17dB MIC Preamp gain setting	570		$mV_{P-P}$	
R <sub>VDAC</sub>	Voice DAC Ripple	300Hz - 3.3kHz through head-phone output.	+/-0.15	+/-0.2	dB (max)	
R <sub>VADC</sub>	Voice ADC Ripple	300Hz - 3.3kHz through head-phone output.	+/-0.25	+/-0.3	dB (max)	
PB <sub>VDAC</sub>	Voice DAC Passband	-3dB Point	3.46		kHz	
SBA <sub>VDAC</sub>	Voice DAC Stopband Attenuation	Above 4kHz	72		dB	
UPB <sub>VADC</sub>	Voice ADC Upper Passband Cutoff Frequency.	Upper -3dB Point	3.47		kHz	

# Electrical Characteristics DV<sub>DD</sub> = 3.3V, AV<sub>DD</sub> = 5V, R<sub>LHP</sub> = 32Ω, R<sub>LHF</sub> = 8Ω 查询"似始9302份"供麻繭ed)

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for  $T_A$ = 25°C.

Symbol	Parameter	Conditions	LM4	1930	Units		
			Typical (Note 6)	Limits (Notes 7, 15)	(Limits)		
LPB <sub>VADC</sub>	Voice ADC Lower Passband Cutoff Frequency.	Lower -3dB Point	0.230		kHz		
SBA <sub>VADC</sub>	Voice ADC Stopband Attenuation	Above 4kHz	65		dB		
SBA <sub>NOTCH</sub>	Voice ADC Notch Attenuation	Centered on 55Hz, figure gives worst case attenuation for 50Hz & 60Hz.	58		dB		
R <sub>DAC</sub>	Audio DAC Ripple	20Hz - 20kHz through head-phone output.	+/-0.1	+/-0.2	dB (max)		
PB <sub>DAC</sub>	Audio DAC Passband Width	-3dB point	22.7		kHz		
SBA <sub>DAC</sub>	Audio DAC Stopband Attenuation	Above 24kHz	76		dB		
DR <sub>DAC</sub>	Audio DAC Dynamic Range Digital Filter Section	Signal = VO at f = 1kHz @ 1% THD+N; f = 1kHz; Noise for -60dBFS digital input; 0dB gain; A-weighted	97		dB		
SNR <sub>DAC</sub>	Audio DAC SNR Digital Filter Section	Signal = VO at f = 1kHz @ 1% THD+N; f = 1kHz; Noise for -60dBFS digital input; 0dB gain; A-weighted	97		dB		
∆A <sub>CH-CH</sub>	Stereo Channel-to-Channel Gain Mismatch		0.3		dB		
V <sub>IL</sub>	Digital Input: Logic Low Voltage Level		0.4		V		
V <sub>IH</sub> [	Digital Input: Logic High Voltage Level		1.4		V		
	Volume Control Range	Maximum Attenuation	-46.5		dB		
	(Headphone amplifiers)	Minimum Attenuation	0		dB		
	Volume Control Range (Mono	Minimum Gain	-34.5		dB		
	speaker amplifier)	Maximum Gain	12		dB		
	Volume Control Step Size (Output amplifiers)		1.5		dB		
	Volume Control Range	Minimum Gain	17		dB		
	(Microphone Preamp)	Maximum Gain	47		dB		
	Volume Control Step Size (Microphone Preamp)		2		dB		
	Side Tone Attenuation Range	Maximum Attenuation	-30		dB		
		Minimum Attenuation	0		dB		
	Side Tone Attenuation Step Size		3		dB		
MCLK	MCLK frequency	CLOCK_DIV = 0 CLOCK_DIV = 1	12.288 24.576		MHz MHz		
	MCLK Duty Cycle		50	40 60	% (min) % (max)		
CONV	Sampling Clock Frequency (Note 9)		48		kHz		
CLKSCL	SCL_CLK Frequency		400		kHz		
RISESCL	SCL_CLK, SCL_DATA Rise Time		300		ns		
FALLSCL	SCL_CLK, SDA_DATA Fall Time		300		ns		
SDAH	SDA_DATA Hold Time		500		ns		
t <sub>SDAS</sub>	SDA_DATA Setup Time		500		ns		

### Electrical Characteristics DV<sub>DD</sub> = 3.3V, AV<sub>DD</sub> = 5V, R<sub>LHP</sub> = 32 $\Omega$ , R<sub>LHF</sub> = 8 $\Omega$ (ல்கூர்'ட்டி M483வர்கி'் 供应商

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for  $T_A = 25$ °C.

Symbol	Parameter	Conditions	LM4930		Units
			Typical	Limits	(Limits)
			(Note 6)	(Notes 7,	
				15)	
f <sub>CLKPCM</sub>	PCM_CLK Frequency	PCM_SYNC_MODE = 00	128		kHz
CLRPCIVI		PCM_SYNC_MODE = 01	256		
		PCM_SYNC_MODE = 10	512		
	PCM_CLK Duty Cycle		50	40	% (min)
				60	% (max)
f <sub>CLKI2S</sub>	I2S_CLK Frequency	I2S_RES = 0	1.536		MHz
		I2S_RES = 1	3.072		
	I2S_CLK Duty Cycle		50	40	% (min)
				60	% (max)

### Electrical Characteristics DV<sub>DD</sub> = 3V, AV<sub>DD</sub> = 3V, R<sub>LHP</sub> = 32 $\Omega$ , R<sub>LHF</sub> = 8 $\Omega$ (Notes 1, 2, 3)

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for  $T_A = 25$ °C.

Symbol	Parameter	Conditions	LM4	1930	Units	
			Typical (Note 6)			
		f <sub>MCLK</sub> = 12.288MHz				
		Output Mode = "0010"				
		Output Mode = "0011"	1.6			
		Output Mode = "0100"				
DI <sub>DD</sub>	Digital Power Supply Current	Output Mode = "0101"				
DIDD	Digital Fower Supply Surrent	Output Mode = "0110"	3.8			
		Output Mode = "0111"				
		Output Mode = "1000"				
		Output Mode = "1001"	4.2	7	mA (max)	
		Output Mode = "1010"				
		f <sub>MCLK</sub> = 12.288MHz; No Load				
		Output Mode = "0010"	5.8			
		Output Mode = "0011"	5.1			
		Output Mode = "0100"	6.5			
Λ1	Analog Power Supply Quiescent	Output Mode = "0101"	6.4			
$AI_{DD}$	Current	Output Mode = "0110"	5.8			
		Output Mode = "0111"	7.0			
		Output Mode = "1000"				
		Output Mode = "1001"	7.5	12	mA (max)	
		Output Mode = "1010"			, ,	
DI <sub>SD</sub>	Digital Powerdown Current	f <sub>MCLK</sub> = 12.288MHz	1	7	μΑ (max)	
-		Output Mode = "0000" Powerdown Mode			,	
Al <sub>SD</sub>	Analog Powerdown Current	f <sub>MCLK</sub> = 12.288MHz			A ( )	
		Output Mode = "0000" Powerdown Mode	0.6	1.5	μA (max)	
DI <sub>ST</sub>	Digital Standby Current	f <sub>MCLK</sub> = 12.288MHz	de 1.1 1.7		ma Λ / \	
-		Output Mode = "0001" Standby Mode			mA (max)	

# Electrical Characteristics DV<sub>DD</sub> = 3V, AV<sub>DD</sub> = 3V, R<sub>LHP</sub> = 32Ω, R<sub>LHF</sub> = 8Ω 查询"似始9302份"供应高ed)

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for  $T_A$ = 25°C.

Symbol	Parameter	Conditions	LM4930		Units
			Typical (Note 6)	Limits (Notes 7, 15)	(Limits)
Al <sub>ST</sub>	Analog Standby Current	f <sub>MCLK</sub> = 12.288MHz Output Mode = "0001" Standby Mode	100	300	μA (max)
$V_{FS\_LS}$	Full-Scale Output Voltage (Mono speaker amplifier)	CLASS = 0; 0dB gain setting; $8\Omega$ BTL load (Note 10)	2.5		$V_{P-P}$
$V_{FS\_HP}$	Full-Scale Output Voltage (Headphone amplifier)	0dB gain setting; $32Ω$ Stereo Load (Note 10)	2.5		$V_{P-P}$
V <sub>MIC_BIAS</sub>	Mic Bias Voltage		2		V
THD+N	Headphone Amplifier Total Harmonic Distortion + Noise	$f_{IN} = 1kHz, P_{OUT} = 7.5mW$	0.07		%
P <sub>OHP</sub>	Headphone Amplifier Output Power	THD+N = 0.5%, $f_{OUT} = 1kHz$	25	15	mW (min)
P <sub>OLS</sub>	Mono Speaker Amplifier Output Power	THD+N = 1%, $f_{OUT}$ = 1kHz	330	270	mW (min)
PSRR	Power Supply Rejection Ratio	$C_{BYPASS} = 1.0 \mu F$ $C_{DAC\_REF} = 1.0 \mu F$ $V_{RIPPLE} = 200 m V_{P-P} @ 217 Hz$	50	42	dB (min)
SNR (Voice)	Signal-to-Noise Ratio (Voice DAC Path)	Signal = Vo at f = 1kHz @1% THD+N, $32\Omega$ Stereo Load; Noise = digital zero, A-weighted; 0dB gain setting	72		dB
SNR (Music)	Signal-to-Noise Ratio (Music Audio Path)	Signal = Vo at f = 1kHz @1% THD+N, 32Ω Stereo Load; Noise = digital zero, A-weighted; 0dB gain setting	86		dB
DR (Voice)	Dynamic Range (Voice DAC Path)	Signal = Vo at f = 1kHz @1% THD+N, 32Ω Stereo Load; Noise for -60dBFS digital input; A-weighted, 0dB gain setting	72		dB
DR (Music)	Dynamic Range (Music Audio Path)	Signal = Vo at f=1kHz @1% THD+N, 32Ω Stereo Load; Noise for -60dBFS digital input; A-weighted, 0dB gain setting	86		dB
SNR <sub>ADC</sub>	Signal-to-Noise Ratio (Voice ADC Path)	Reference signal = 0dBFS MIC_P, MIC_N terminated with 10Ω to ground; A-weighted; 47dB MIC preamp gain setting	75		dB
DR <sub>ADC</sub>	Dynamic Range (Voice ADC Path)	Reference signal = 0dBFS Noise for -60dBFS digital input; A-weighted; 47dB MIC preamp gain setting	75		dB
X <sub>TALK</sub>	Stereo Channel-to-Channel Crosstalk	$f_S = 48kHz$ , $f_{IN} = 1kHz$ sinewave at $-3dB_{FS}$	73		dB
V <sub>MIC-IN</sub>	Maximum Differential MIC Input Voltage	17dB MIC Preamp gain setting	570		$mV_{P-P}$
R <sub>VDAC</sub>	Voice DAC Ripple	300Hz - 3.3kHz through head-phone output.	+/-0.15	+/-0.2	dB (max)
R <sub>VADC</sub>	Voice ADC Ripple	300Hz - 3.3kHz through head-phone output.	+/-0.25	+/-0.3	dB (max)
PB <sub>VDAC</sub>	Voice DAC Passband	-3dB Point	3.46		kHz

## Electrical Characteristics DV<sub>DD</sub> = 3V, AV<sub>DD</sub> = 3V, R<sub>LHP</sub> = 32 $\Omega$ , R<sub>LHF</sub> = 8 $\Omega$ (No臺询"共例483 $\Omega$ ii) 从中达商

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for  $T_A = 25^{\circ}C$ .

Symbol	Parameter	Conditions	LM4	4930	Units (Limits)
			Typical (Note 6)	Limits (Notes 7, 15)	
BA <sub>VDAC</sub>	Voice DAC Stopband Attenuation	Above 4kHz	72		dB
PB <sub>VADC</sub>	Voice ADC Upper Passband Cutoff Frequency.	Upper -3dB Point	3.47		kHz
PB <sub>VADC</sub>	Voice ADC Lower Passband Cutoff Frequency.	Lower -3dB Point	0.230		kHz
BA <sub>VADC</sub>	Voice ADC Stopband Attenuation	Above 4kHz	65		dB
BA <sub>NOTCH</sub>	Voice ADC Notch Attenuation	Centered on 55Hz, figure gives worst case attenuation for 50Hz & 60Hz.	58		dB
DAC	Audio DAC Ripple	20Hz - 20kHz through head-phone output.	+/-0.1	+/-0.2	dB (max)
B <sub>DAC</sub>	Audio DAC Passband Width	-3dB point	22.7		kHz
BA <sub>DAC</sub>	Audio DAC Stopband Attenuation	Above 24kHz	76		dB
R <sub>DAC</sub>	Audio DAC Dynamic Range Digital Filter Section	Signal = VO at f = 1kHz @ 1% THD+N; f = 1kHz; Noise for -60dBFS digital input; 0dB gain; A-weighted	97		dB
NR <sub>DAC</sub>	Audio DAC SNR Digital Filter Section	Signal = VO at f = 1kHz @ 1% THD+N; f = 1kHz; Noise for -60dBFS digital input; 0dB gain; A-weighted	97		dB
A <sub>CH-CH</sub>	Stereo Channel-to-Channel Gain Mismatch		0.3		dB
'IL	Digital Input: Logic Low Voltage Level		0.4		V
'IH	Digital Input: Logic High Voltage Level		1.4		V
	Volume Control Range	Maximum Attenuation	-46.5		dB
	(Headphone amplifiers)	Minimum Attenuation	0		dB
	Volume Control Range (Mono	Minimum Gain	-34.5		dB
	speaker amplifier)	Maximum Gain	12		dB
	Volume Control Step Size (Output amplifiers)		1.5		dB
	Volume Control Range	Minimum Gain	17		dB
	(Microphone Preamp)	Maximum Gain	47		
	Volume Control Step Size (Microphone Preamp)		2		dB
	Side Tone Attenuation Range	Maximum Attenuation	-30		dB
	0.1 7 4	Minimum Attenuation	0		dB
	Side Tone Attenuation Step Size		3		dB
ICLK	MCLK frequency	CLOCK_DIV = 0 CLOCK_DIV = 1	12.288 24.576		MHz MHz
	MCLK Duty Cycle		50	40 60	% (min) % (max)
ONV	Sampling Clock Frequency	(Note 9)	48		kHz
LKSCL	SCL_CLK Frequency		400		kHz
ISESCL	SCL_CLK, SCL_DATA Rise Time		300		ns
ALLSCL	SCL_CLK, SDA_DATA Fall Time		300		ns
SDAH	SDA_DATA Hold Time		500		ns

### Electrical Charac 算词"(M4930209)"供应商ed) Electrical Characteristics DV<sub>DD</sub> = 3V, AV<sub>DD</sub> = 3V, R<sub>LHP</sub> = 32 $\Omega$ , R<sub>LHF</sub> = 8 $\Omega$

The following specifications apply for the circuit shown in Figure 1, unless otherwise specified. Limits apply for TA= 25°C.

Symbol	Parameter	Conditions	LM4930		Units
			Typical (Note 6)	Limits (Notes 7, 15)	(Limits)
t <sub>SDAS</sub>	SDA_DATA Setup Time		500		ns
f <sub>CLKPCM</sub>	PCM_CLK Frequency	PCM_SYNC_MODE = 00	128		kHz
		PCM_SYNC_MODE = 01	256		kHz
		PCM_SYNC_MODE = 10	512		kHz
	DCM CLK Duty Cycle		50	40	% (min)
	PCM_CLK Duty Cycle		50	60	% (max)
f <sub>CLKI2S</sub>	I2S_CLK Frequency	I2S_RES = 0	1.536		MHz
		I2S_RES = 1	3.072		MHz
	ISS CLK Duty Cycle		50	40	% (min)
	I2S_CLK Duty Cycle		50	60	% (max)

Note 1: Absolute Maximum Ratings indicate limits beyond which damage to the device may occur. Operating Ratings indicate conditions for which the device is functional but do not guarantee specific performance limits. Electrical Characteristics state DC and AC electrical specifications under particular test conditions which guarantee specific performance limits. This assumes that the device is within the Operating Ratings. Specifications are not guaranteed for parameters where no limit is given, however, the typical value is a good indication of device performance.

Note 2: All voltages are measured with respect to the relevant GND pin unless otherwise specified.

Note 3: The maximum power dissipation must be derated at elevated temperatures and is dictated by T<sub>JMAX</sub>,  $\theta_{JA}$ , and the ambient temperature, T<sub>A</sub>. The maximum allowable power dissipation is  $P_{DMAX} = (T_{JMAX} - T_A) / \theta_{JA}$  or the number given in Absolute Maximum Ratings, whichever is lower. For the LM4930, see power derating currents for more information.

Note 4: Human body model: 100pF discharged through a  $1.5k\Omega$  resistor.

Note 5: Machine model: 220pF - 240pF discharged through all pins.

Note 6: Typicals are measured at 25°C and represent the parametric norm.

Note 7: Limits are guaranteed to National's AOQL (Average Outgoing Quality Level).

Note 8: Best operation is achieved by maintaining 3.0V ≤ AV<sub>DD</sub> ≤ 5.0 and 3.0V ≤ DV<sub>DD</sub> ≤ 3.6V. AV<sub>DD</sub> must be equal to or greater than DV<sub>DD</sub>. for proper operation.

Note 9: The sampling clock frequency is equal to the master clock frequency divided by 256. (f<sub>conv</sub> = f<sub>MCLK</sub>/256)

Note 10: This value represents the 0dB output level of the given amplifier for the given analog supply voltage. Gain values given in the GAINCONFIG register are relative to these full-scale values for each output amplifier.

Note 11: To ensure a successful transistion into Powerdown Mode, ZXD\_DISABLE must be set whenever there is no audio input signal present.

Note 12: It is recommended to alter this bit only while the part is in Powerdown Mode.

Note 13: Reserved bits should be set to zero when programming the associated register.

Note 14: With the exception of Standby Mode, rapid switching between modes should be avoided. Rapid switching between modes will not ensure that the desired mode will be activated.

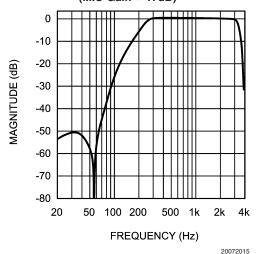
Note 15: Datasheet min/max specification limits are guaranteed by design, test, or statistical analysis.

Note 16: 0dBm0 = -3dBFS for the PCM voice codec and 0dBm0 = -1dBFS for the I<sup>2</sup>S DAC, unless otherwise specified.

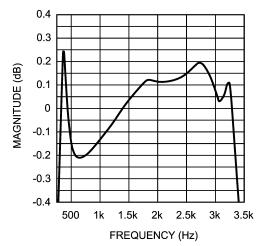
Note 17: The given θ<sub>Δ</sub> is for an LM4930 packaged in an LQA44A with the Exposed-DAP soldered to an exposed 2in<sup>2</sup> area of 1oz printed circuit board copper with 16 thermal vias as described in National AN-1187.

## Typical Performance Characteristics (Note 16) 查询 MC Preamp 1 Aborteristics (Note 16) MI

(MIC Gain = 17dB)

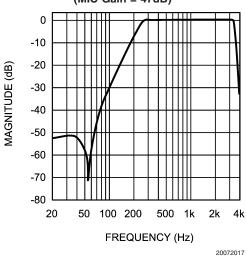


MIC PreAmp + ADC Frequency Response Zoom (MIC Gain = 17dB)

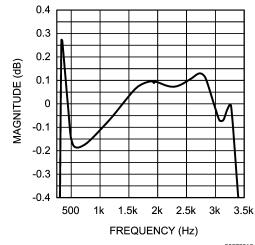


20072016

MIC PreAmp + ADC Frequency Response (MIC Gain = 47dB)

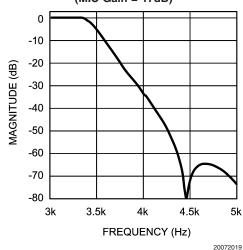


MIC PreAmp + ADC Frequency Response Zoom (MIC Gain = 47dB)

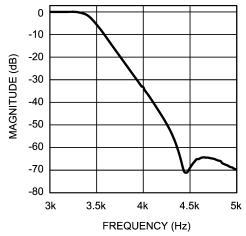


20072018

MIC PreAmp + ADC Frequency Response High Cutoff (MIC Gain = 17dB)



MIC PreAmp + ADC Frequency Response High Cutoff (MIC Gain = 47dB)

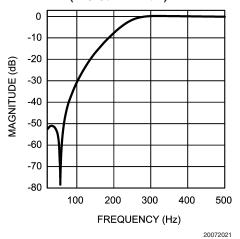


20072020

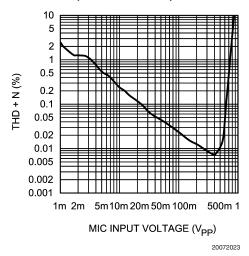
### Typical Performance Characteristics (Note 16) (Continued)

询"LM4930\_04"供应商

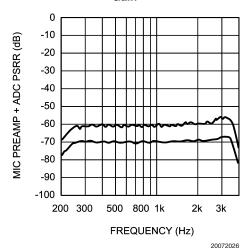
MIC PreAmp + ADC Frequency Response Low Cutoff (MIC Gain = 17dB)



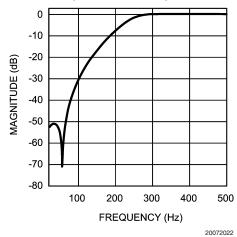
ADC THD+N vs MIC Input Voltage (MIC Gain = 17dB)



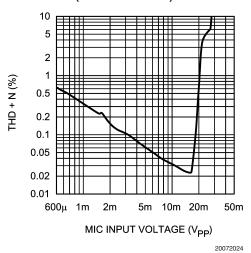
MIC PreAmp + ADC PSRR vs Frequency
Top Trace = 47dB MIC Gain, Bottom Trace = 17dB MIC
Gain



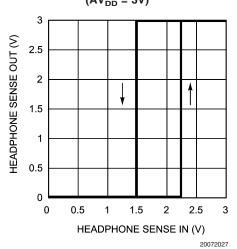
MIC PreAmp + ADC Frequency Response Low Cutoff (MIC Gain = 47dB)



ADC THD+N vs MIC Input Voltage (MIC Gain = 47dB)

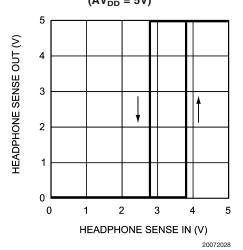


Headphone Sense In Hysteresis Loop  $(AV_{DD} = 3V)$ 

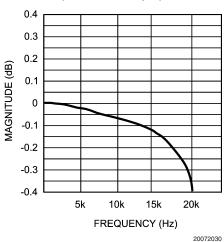


### Typical Performance Characteristics (Note 16) (Continued) 查询"LM4930\_04"供应商

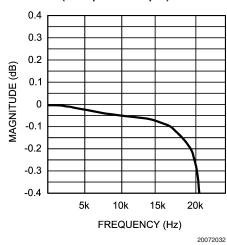
### Headphone Sense In Hysteresis Loop $(AV_{DD} = 5V)$



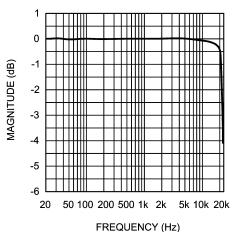
### I<sup>2</sup>S DAC Frequency Response Zoom (Handsfree Output)



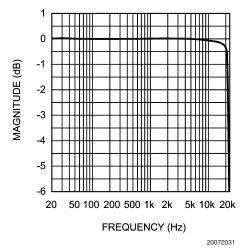
### I<sup>2</sup>S DAC Frequency Response Zoom (Headphone Output)



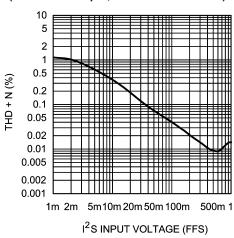
### I<sup>2</sup>S DAC Frequency Response ( Handsfree Output)



### I<sup>2</sup>S DAC Frequency Response Zoom (Headphone Output)



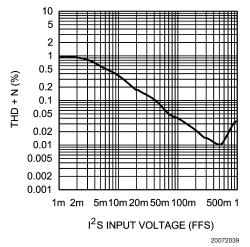
THD+N vs I2S Input Voltage (Handsfree Output, 0dB Handsfree Gain)



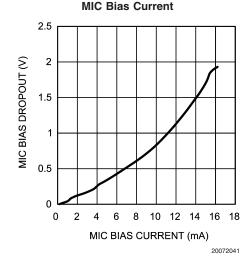
20072038

### Typical Perform july 1.00 minus Typical Performance Characteristics (Note 16) (Continued)

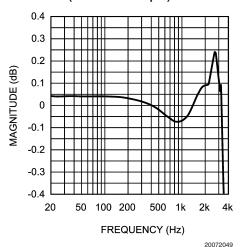
THD+N vs I2S Input Voltage (Headphone Output, 0dB Headphone Gain)



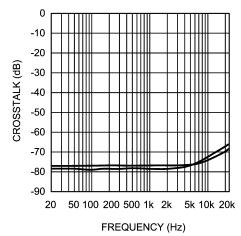
MIC Bias Dropout Voltage vs



**PCM DAC Frequency Response Zoom** (Handsfree Output)

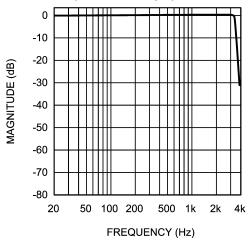


I2S DAC Crosstalk (Top Trace = Left to Right, Bottom Trace = Right to Left)



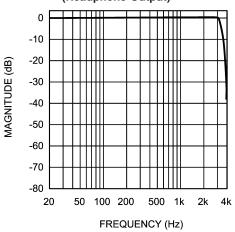
20072040

**PCM DAC Frequency Response** (Handsfree Output)



20072055

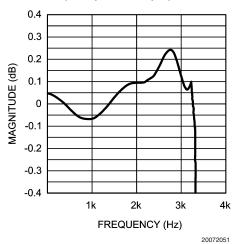
**PCM DAC Frequency Response** (Headphone Output)



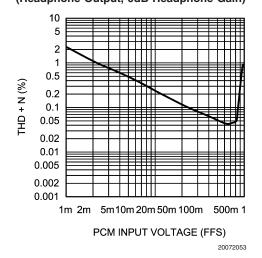
20072050

### Typical Performance Characteristics (Note 16) (Continued) 查询"LM4930\_04"供应商

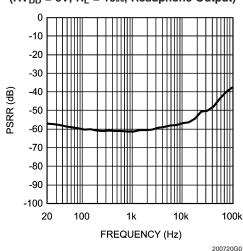
PCM DAC Frequency Response Zoom (Headphone Output)



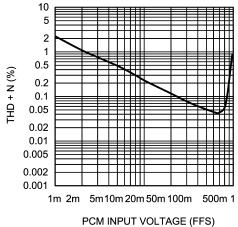
THD+N vs PCM Input Voltage (Headphone Output, 0dB Headphone Gain)



**PSRR** vs Frequency  $(AV_{DD} = 3V, R_L = 16\Omega, Headphone Output)$ 

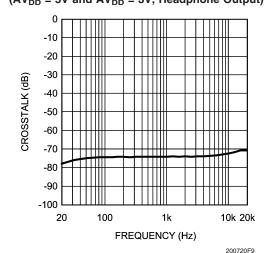


THD+N vs PCM Input Voltage (Handsfree Output, 0dB Handsfree Gain)

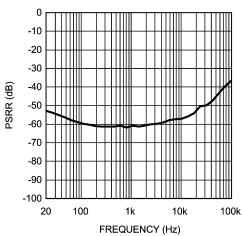


20072052

Crosstalk  $(AV_{DD} = 5V \text{ and } AV_{DD} = 3V, \text{ Headphone Output})$ 



**PSRR vs Frequency**  $(AV_{DD} = 3V, R_L = 32\Omega, Headphone Output)$ 

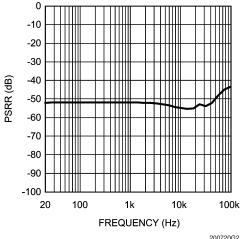


200720G1

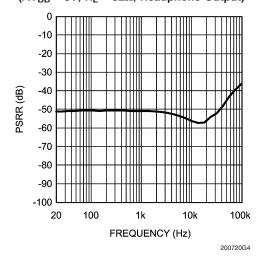
### Typical Performance Characteristics (Note 16) (Continued)

询"LM4930\_04"供应商

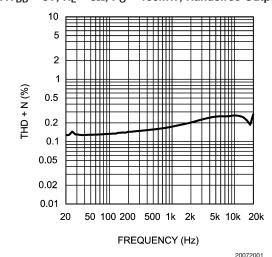
**PSRR** vs Frequency  $(AV_{DD} = 3V, R_L = 8\Omega, Handsfree Output)$ 



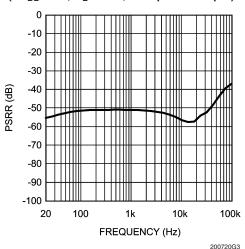
**PSRR** vs Frequency (AV<sub>DD</sub> = 5V,  $R_L$  = 32 $\Omega$ , Headphone Output)



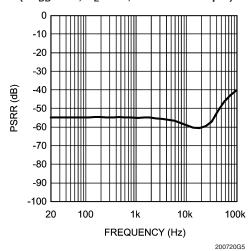
THD+N vs Frequency  $(AV_{DD} = 3V, R_L = 8\Omega, P_O = 150mW, Handsfree Output)$ 



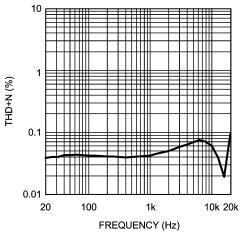
**PSRR vs Frequency** (AV<sub>DD</sub> = 5V,  $R_L$  = 16 $\Omega$ , Headphone Output)



**PSRR** vs Frequency (AV<sub>DD</sub> = 5V,  $R_L$  = 8 $\Omega$ , Handsfree Output)



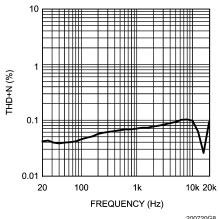
THD+N vs Frequency (AV $_{DD}$  = 5V and AV $_{DD}$  = 3V, R $_{L}$  = 16 $\Omega$ , P $_{O}$  = 15mW, **Headphone Output)** 



200720G7

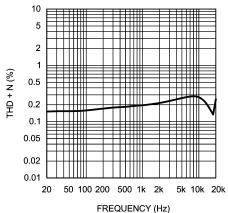
### Typical Performance Characteristics (Note 16) (Continued) 查询"LM4930\_04"供应商

THD+N vs Frequency (AV $_{DD}$  = 5V and AV $_{DD}$  = 3V, R $_{L}$  = 32 $\!\Omega,$  P $_{O}$  = 7.5mW, **Headphone Output)** 

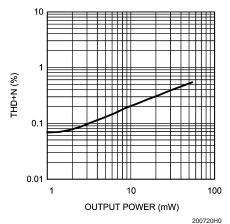


200720G8

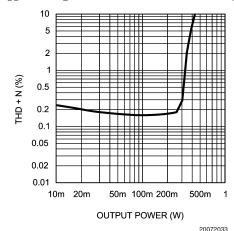
THD+N vs Frequency  $(AV_{DD} = 5V, R_L = 8\Omega, P_O = 250mW, Handsfree Output)$ 



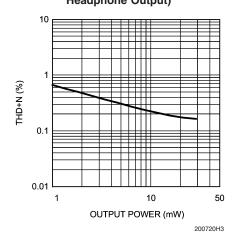
THD+N vs Output Power  $AV_{DD} = 3V$ ,  $R_L = 16\Omega$ , f = 1kHz, Headphone Output)



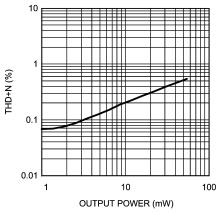
THD+N vs Output Power  $(AV_{DD} = 3V, R_L = 8\Omega, f = 1kHz, Handsfree Output)$ 



THD+N vs Output Power (AV $_{\rm DD}$  = 5V and AV $_{\rm DD}$  = 3V, R $_{\rm L}$  = 32 $\!\Omega,$  f = 1kHz, **Headphone Output)** 



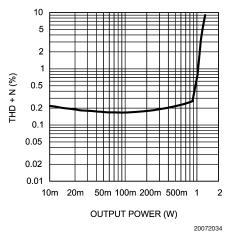
THD+N vs Output Power (AV $_{\rm DD}$  = 5V and AV $_{\rm DD}$  = 3V, R $_{\rm L}$  = 16 $\Omega,$  f = 1kHz, **Headphone Output)** 



200720H4

### Typical Performance Characteristics (Note 16) (Continued) 询"LM4930\_04"供应商

THD+N vs Output Power  $(AV_{DD} = 5V, R_L = 8\Omega, f = 1kHz, Handsfree Output)$ 



### **Application Information**

REFERENCE DESIGN BOARD AND LAYOUT

LM4930ITL Board Layout

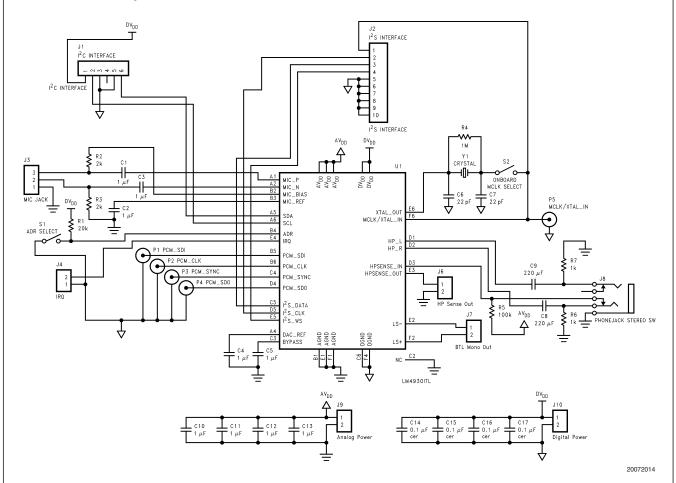


FIGURE 5. LM4930ITL Demo Board Schematic

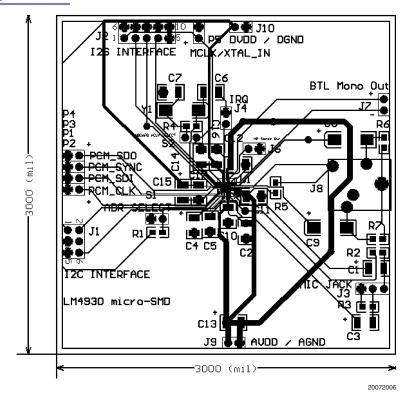


FIGURE 6. LM4930ITL Demo Board Composite View

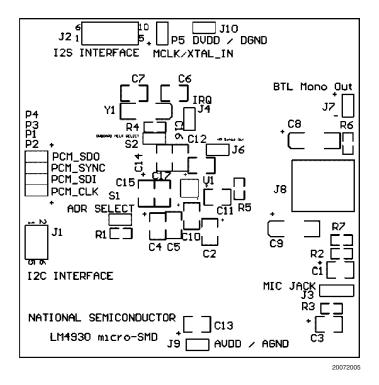
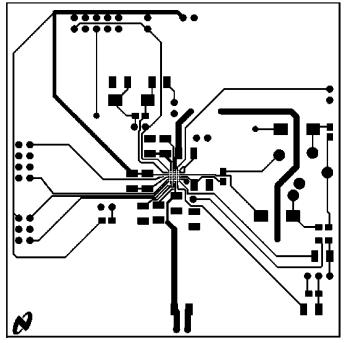


FIGURE 7. LM4930ITL Demo Board Silkscreen



20072004

FIGURE 8. LM4930ITL Demo Board Top Layer

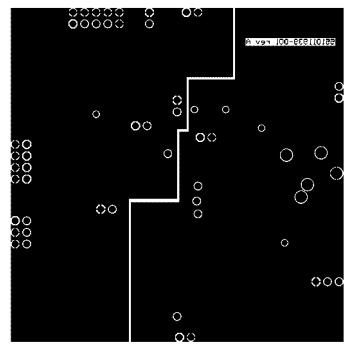
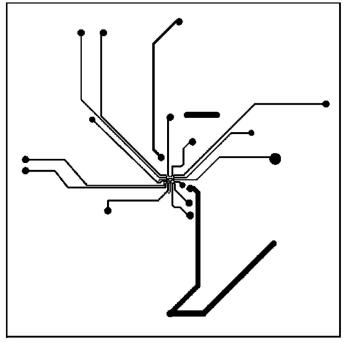
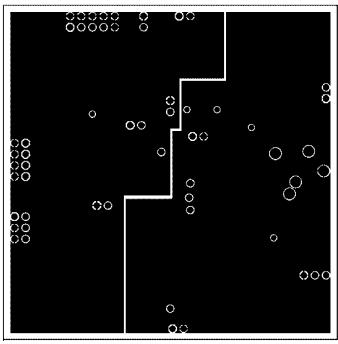


FIGURE 9. LM4930ITL Demo Board Bottom Layer



20072012

FIGURE 10. LM4930ITL Demo Board Inner Layer 1



20072011

FIGURE 11. LM4930ITL Demo Board Inner Layer 2

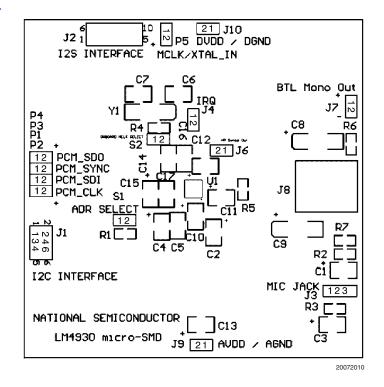


FIGURE 12. Pin Markings for LM4930ITL demo board

**BILL OF MATERIALS FOR LM4930** 

### LM4930 Demo Board Bill Of Materials

Comment	Footprint	Designators			
1k	0805	R6, R7			
2k	0805	R2, R3			
20k	0805	R1			
100k	0805	R5			
1M	0805	R4			
22pF	1210	C6, C7			
0.01µF cer	1210	C16, C17			
0.1µF cer	1210	C14, C15			
1μF	1210	C1, C2, C3, C4, C5, C10, C11,			
		C12, C13			
220µF	7243	C8, C9			
CRYSTAL	7243	Y1			
PHONE JACK STEREO SW STE	PHONE JACK STEREO SW STEREO HEADPHONE JACK (3.5MM) J8				

### Two-wire control Interface (J1)

Pin	Function
1	DVDD
2	SCL
3	DGND
4	NC
5	DGND
6	SDA

### PCM Interface (P4, P3, P1, P2)

Header	Function
P1	PCM_SDI
P2	PCM_CLK
P3	PCM_SYNC
P4	PCM_SDO

### I2S Interface (J2)

Pin	Function
1	MCLK
2	I2S-CLK
3	I2S-DATA
4	I2S-WS
5	DGND
6	DGND
7	DGND
8	DGND
9	DGND
10	DGND

### MIC Jack

Pin	Function
1	AGND
2	MIC-
3	MIC+

### **Misc Jumpers and Headers** DVDD/DGND (J10)

Pin	Function
1	DGND
2	AVDD

### **Misc Jumpers and Headers** AVDD/AGND (J9)

Pin	Function
1	AGND
2	AVDD

### Misc Jumpers and Headers MCLK/XTAL\_IN (P5)

Pin	Function

### **Application Information** (Continued)

询"LM4930\_04"供应商

### Misc Jumpers and Headers MCLK/XTAL\_IN (P5) (Continued)

1	DGND
2	MCLK/XTAL_IN

### ADR SELECT (S1)

Jumper IN = LOW

Control interface responds to addresses 001000b (BASICCONFIG), 0010001b (VOICETESTCONFIG)), and 0010010b (GAIN-CONFIG)

Jumper OUT = HIGH

Control interface responds to addresses 111000b (BASICCONFIG), 1110001b (VOICETESTCONFIG)), and 1110010b (GAIN-CONFIG)

### HP Sense Out (J6)

Pin	Function
1	AGND
2	HPSense_Out

### **IRQ (J4)**

Pin	Function
1	DGND
2	IRQ

#### Onboard MCLK Select (S2)

Jumper IN = Onboard MCLK

Jumper OUT = External MCLK

### LM4930ITL DEMO BOARD OPERATION

The LM4930ITL demo board is a complete evaluation platform, designed to give easy access to the control pins of the part and comprise all the necessary external passive components. Besides the separate analog (J9) and digital (J10) supply connectors, the board features seven other major input and control blocks: a two wire interface bus (J1) for the control lines, a PCM interface bus (P1-P4) for voiceband digital audio, an I2S interface bus (J2) for full-range digital audio, an analog mic jack input (J3) for connection to an external microphone, a BTL mono output (J7) for connection to an external speaker, a stereo headphone output (J8), and an external MCLK input (P5) for use in place of the crystal on the demoboard.

### Two-wire Interface Bus (J1)

This is the main control bus for the LM4930. It is a two-wire interface with an SDA line (data) and SCL line (clock). Each transmission from the baseband controller to the LM4930 is given MSB first and must follow the timing intervals given in the Electrical Characteristics section of the datasheet to create the start and stop conditions for a proper transmission. The start condition is detected if SCL is high on the falling edge of SDA. The stop condition is detected if SCL is high on the rising edge of SDA. Repeated start signals are handled correctly. Data is then transmitted as shown in Figure 2. After the start condition has been achieved the chip address is sent, followed by a set write bit, wait for ack (SDA will be pulled low by LM4930), data bits 15-8, wait for ACK (SDA will be pulled low by LM4930), data bits 7-0, wait for ACK (SDA will be pulled low by LM4930) and finally the stop condition is given.

This same sequence follows for any control bus transmission to the LM4930. The chip address is hardwire selected by the ADR Select pin which may be jumpered high or low with its application at S1 on the demo board. The chip address is then given as a combination of the identifying bits for the LM4930 plus the 2-bit address of the desired control register (00b = BasicConfig, 01b = VoicetestConfig, 10b = GainConfig). Acceptable addresses are shown here in Table 1.

Table 1. LM4930 Control Bus Addresses

Address Bits					Register Address			
ADR	= 0							
6	5	4	3	2	1	0		
0	0	1	0	0	0	0		
0	0	1	0	0	0	1		
0	0	1	0	0	1	0		
ADR = 1								
1	1	1	0	0	0	0		
1	1	1	0	0	0	1		
1	1	1	0	0	1	0		

Data is sampled only if the address is in range and the  $R/\overline{W}$  bit is clear. Data for each register is given in the System Control Registers section of the datasheet. National Semi-conductor also features a special control board for quick evaluation of the LM4930 demo board with your PC. This is a serial control interface board, complete with header com-

### **Application Information** (Continued)

patie vith the interface header (京) on the LM4930 board. This also features demonstration software to allow for complete control and evaluation of the various modes and functions of the LM4930 through the bus.

Pullup resistors are required to achieve reliable operation.  $750\Omega$  pullup resistors on the SDA and SCL lines achieves best results when used with National's parallel-to-serial interface board. Lower value pullup resistors will decrease the rise and fall times on the bus which will in turn decrease susceptibility to bus noise that may cause a false trigger. The cost comes at extra current use. Control bus reliability will thus depend largely on bus noise and may vary from design to design. Low noise is critical for reliable operation.

### PCM Bus Interface (P1, P2, P3, P4)

PCM\_SDO (P4), PCM\_SYNC (P3), PCM\_SDI (P1), and PCM\_CLK (P2) form the PCM interface bus for simple communication with most baseband ICs with voiceband communications and follow the PCM-1900 communications standard. The PCM interface features frame lengths of 16, 32, or 64 bits, A-law and u-law companding, linear mode, short or long frame sync, an energy-saving power down mode, and master only operation.

The PCM bus does not support a slave mode. It operates as a master only. Thus PCM\_SYNC and PCM\_CLK are solely generated by the LM4930. PCM\_SYNC is the word sync line for the bus. It operates at a fixed frequency of 8kHz and may be set in the BASICCONFIG register (bit 5 PCM\_LONG) for short or long frame sync. A short frame sync is 1 PCM\_CLK cycle (PCM\_LONG=0), a long frame sync is 2 PCM\_CLK cycles long (PCM\_LONG=1). A long sync pulse is also delayed one clock cycle relative to a short sync pulse. This is illustrated in Figure 3. PCM\_CLK is the bit clock for the bus. It's frequency depends on the number of 16-bit frames per sync pulse and can be 128kHz, 256kHz, 512kHz.

The other two lines, PCM\_SDO and PCM\_SDI, are for serial data out and serial data in, respectively. The type of data may also be set in the BASICCONFIG register by bits 6 and 7. Bit 6 controls whether the data is linear or companded. If set to 1, the 8 MSBs are presumed to be companded data and the 8 LSBs are ignored. If cleared to 0, the data is treated as 2's complement PCM data. Bit 7 controls which PCM law is used if Bit 6 is set for companded (G711) data. If set to 1, the companded data is assumed to be A-law. If cleared to 0, the companded data is treated as µ-law.

Bits 8:9 of the BASICCONFIG register set the PCM\_SYNC\_MODE settings. This controls the number of 16 bit frames per sync pulse. The feature allows the LM4930 to function harmoniously with other devices or channels on the PCM bus by adjusting the number of 16 bit frames per sync pulse to 1 (00b), 2 (01b), or 4 (10b). The LM4930 will transmit PCM data in the first frame and then tri-state the PCM\_SDO pin on later frames.

In addition, the LM4930 provides control to allow the PCM\_CLK and PCM\_SYNC clocks to continue functioning even when the LM4930 is in Standby mode. By setting bit 10 of the BASICCONFIG register to 1 PCM\_ALWAYS\_ON is enabled and the LM4930 will continue to drive the PCM clock and sync lines when in Standby mode. This bit should be set if another codec is using the PCM bus. Powerdown mode will disable these outputs.

#### I2S Interface Bus (J2)

The I2S standard provides a uni-directional serial interface designed specifically for digital audio. For the LM4930, the interface provides access to a 48kHz, 16 bit full-range stereo audio DAC. This interface uses a three port system of clock (I2S\_CLK), data (I2S\_DATA), and word (I2S\_WS). The clock and word lines can be either master or slave as set by bit 11 in the BASICCONFIG register.

A bit clock (I2S\_CLK) at 32 or 64 times the sample frequency is established by the I2S system master and a word select (I2S\_WS) line is driven at a frequency equal to the sampling rate of the audio data, in this case 48kHz. The word line is registered to change on the negative edge of the bit clock. The serial data (I2S\_DATA) is sent MSB first, again registered on the negative edge of the bit clock, delayed by 1 bit clock cycle relative to the changing of the word line (typical I2S format - see Figure 4).

The resolution of the I2S interface may be set by modifying the I2S\_RES bit (bit 12) in the BASICCONFIG register. If set to 1, the LM4930 operates at 32 bits per frame (3.072MHz). If cleared to 0, then 16 bits per frame is selected (1.536MHz). This has a corresponding effect on the bit clock. The I2S Interface Bus also provides for an additional MCLK connection to an external device from the LM4930 demo board. This may be used in conjunction with National Semiconductors SPDIF->I2S Conversion Board for quick evaluation. This board features a connection header that interfaces with pins 1-5 of the I2S Interface Bus. Pins 6-10 are provided as digital ground references for the case of discrete connections.

### MCLK/XTAL\_IN (P5)

This is the input for an external Master Clock. The jumper at S2 must be removed (disconnecting the onboard crystal from the circuit) when using an external Master Clock.

### BTL Mono Out (J7)

This is the mono speaker output, designed for use with an 8 ohm speaker. The outputs are driven in bridge-tied-load (BTL) mode, so both sides have signal. Outputs are normally biased at one half AVDD when the LM4930 is in active mode. Additionally, if the CLASS bit is set to 1 in the VOICETEST-CONFIG register (bit 0) the BTL mono output is internally configured as a buffer amplifier designed for use with an external class D amp.

### Stereo Headphone Out (J8)

This is the stereo headphone output. Each channel is singleended, with 220uF DC blocking capacitors mounted on the demo board. The jack features a typical stereo headphone pinout.

A headphone sense pin is provided at J6. This pin provides a clean logic high or low output to indicate the presence of headphones in the headphone jack. A common application circuit for this is given in the Reference Board Schematic shown in Figure 5. In this application HPSENSE\_IN is pulled low by the 1k ohm resistor when no headphone is present. This gives a corresponding logic low output on the HPSENSE\_OUT pin. When a headphone is placed in the jack the 1k ohm pull-down is disconnected and a 100k ohm pull-up resistor creates a high voltage condition on HPSENSE\_IN. This in turn creates a logic high on HPSENSE\_OUT. This output may be used to reliably drive an external microcontroller with headphone status.

### Application Info **Application Information** (Continued)

This jack is for connection to an external microphone like the kind typically found in mobile phones. Pin 1 is GND, pin 2 is the negative input pin, and pin 3 is the positive pin, with phantom voltage supplied by MIC\_BIAS on the LM4930.

#### **IRQ (J4)**

This pin provides simple status updates from the LM4930 to an external microcontroller if desired. IRQ is logic high when the LM4930 is in a stable state and changes to low when changing modes. This can also be useful for simple software/driver development to monitor mode changes, or as a simple debugging tool.

#### **BASIC OPERATION**

The LM4930 is a highly integrated audio subsystem with many different operating modes available. These modes may be controlled in the BASICCONFIG register in bits 3:0. These mode settings are shown in the BASICCONFIG register table and are described here below:

#### Powerdown Mode (0000b)

Part is powered down, analog outputs are not biased. This is a minimum current mode. All part features are shut down.

#### Standby Mode (0001b)

The LM4930 is powered down, but outputs are still biased at one half AVDD. This comes at some current cost, but provides a much faster turn-on time with zero "click and pop" transients on the headphone out. Standby mode can be toggled into and out of rapidly and is ideal for saving power whenever continuous audio is not a requirement. All other part functions are suspended unless PCM ALWAYS ON (bit 10 in BASICCONFIG register) is enabled, in which case PCM\_CLK and PCM\_SYNC will continue to function.

### Mono Speaker Mode (0010b)

Part is active. All analog outputs are biased. Audio from the voiceband codec is routed to the mono speaker out. Stereo headphone out is silent.

### Headphone Call Mode (0011b)

Part is active. All analog outputs are biased. Audio from voiceband codec is routed to the stereo headphones. Both left and right channels are the same. Mono speaker out is silent.

### Conference Call Mode (0100b)

Part is active. All analog outputs are biased. Audio from the voiceband codec is routed to the mono speaker out and to the stereo headphones.

### L+R Mixed to Mono Speaker (0101b)

Part is active. All analog outputs are biased. Full-range audio from the 16bit/48kHz audio DAC is mixed together and routed to the mono speaker out. Stereo headphones are silent.

### Headphone Stereo Audio (0110b)

Part is active. All analog outputs are biased. Full-range audio from the 16bit/48kHz audio DAC is sent to the stereo headphone jack. Each channel is heard discretely. The mono speaker is silent.

### L+R Mixed to Mono Speaker + Stereo Headphone Audio (0111b)

Part is active. All analog outputs are biased. Full-range audio from the 16bit/48kHz audio DAC is sent discretely to the stereo headphone jack and also mixed together and sent to the mono speaker out.

#### Mixed Mode (1000b)

Part is active. All analog outputs are biased. This provides one channel (the left channel) of full range audio to the mono speaker out. Audio from the voiceband codec is then sent to the stereo headphones, the same on each channel.

#### Mixed Mode (1001b)

Part is active. All analog outputs are biased. Mixed voiceband and full-range audio (left channel only) is sent to the mono speaker out. Audio from the voiceband codec only is sent to the stereo headphones, the same on each channel.

#### Mixed Mode (1010b)

Part is active. All analog outputs are biased. Audio from the voiceband codec is sent to the mono speaker out. The left channel only of the full range audio is then sent to both the left and right channels of the stereo headphone out.

#### **REGISTERS**

The LM4930 starts on power-up with all registers cleared in Powerdown mode. Powerdown mode is the recommended time to make setup changes to the digital interfaces (PCM bus, I2S bus). Although the configuration registers can be changed in any mode, changes made during Standby or Powerdown prevent unwanted audio artifacts that may occur during rapid mode changes with the outputs active. The LM4930 also features a soft reset. This reset is enabled by setting bit 4 of the BASICCONFIG register.

The VOICETESTCONFIG register is used to set various configuration parameters on the voiceband and full-range audio codecs. SIDETONE\_ATTEN (bits 4:1) refers to the level of signal from the MIC input that is fed back into the analog audio output path (commonly used in headphone applications and killed in hands-free applications). Setting the AUTOSIDE bit (bit 5) automatically mutes the sidetone in voice over mono speaker modes so feedback isn't an issue.

Quick mute functions are also located in this register, with bits 13:15 muting the mono speaker amp, the headphone amp, and the mic preamp respectively.

This register also has a CLOCK\_DIV bit (bit 6) which, if set, allows for the use of a 24.576MHz clock instead of the default 12.288MHz.

The GAINCONFIG register is used to control the gain of the mono speaker amp, the headphone amp, and the mic preamp. This allows flexible mono speaker gains from -34.5dB to +12dB in 1.5dB steps, headphone amp gains of -46.5dB to 0dB in 1.5dB steps, and mic preamp gains of 17dB to 47dB in 2dB steps. Gain levels may be modified in any mode, but may wait for a zero cross detect in the DAC to eliminate volume control artifacts. This wait for zero cross may be disabled by setting the ZXD\_DISABLE bit (bit 7) in the VOICETESTCONFIG register to allow immediate changes.

### **ANALOG INPUTS AND OUTPUTS**

The LM4930 features an analog mono BTL output for connection to an  $8\Omega$  external speaker. This output can provide up to 1W of power into an  $8\Omega$  load with a 5V analog supply.

Application Information (Continued)
A 意识 M4930 04 供应商 output is also featured, providing up to 30mW of power per channel into  $32\Omega$  with a 5V analog supply.

A Headphone Sense output is provided on J6 for connection to an external controller. This pin goes high when a heaphone is present (when used as shown in Figure 5) and will function in all modes independent of other operations the LM4930 may be currently processing.

The MIC Jack input (J3) provides for a low level analog input. Pin 3 provides the power to the MIC and the positive input of the LM4930. Gain for the MIC preamp is set in the GAIN-CONFIG register.

36-Bump micro SMD Order Number LM4930ITL NS Package Number TLA36KRA  $X_1 = 3.230 \pm 0.03$ mm  $X_2 = 3.408 \pm 0.03$   $X_3 = 0.600 \pm 0.075$ 

 $36 \times 0^{0.335}$ 0.305 À

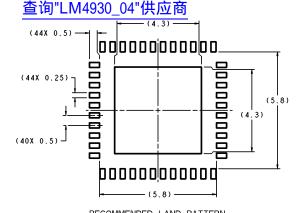
TLA36XXX (Rev C)

SILICON-

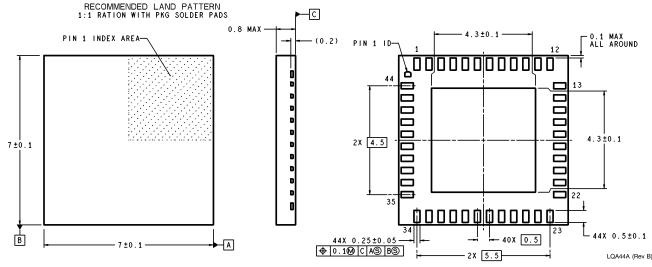
www.national.com 3

-BUMP A1 CORNER

### Physical Dimensions inches (millimeters) unless otherwise noted (Continued)



DIMENSIONS ARE IN MILLIMETERS



LLP Package Order Number LM4930LQ NS Package Number LQA44A

### LIFE SUPPORT POLICY

NATIONAL'S PRODUCTS ARE NOT AUTHORIZED FOR USE AS CRITICAL COMPONENTS IN LIFE SUPPORT DEVICES OR SYSTEMS WITHOUT THE EXPRESS WRITTEN APPROVAL OF THE PRESIDENT AND GENERAL COUNSEL OF NATIONAL SEMICONDUCTOR CORPORATION. As used herein:

- Life support devices or systems are devices or systems which, (a) are intended for surgical implant into the body, or (b) support or sustain life, and whose failure to perform when properly used in accordance with instructions for use provided in the labeling, can be reasonably expected to result in a significant injury to the user.
- A critical component is any component of a life support device or system whose failure to perform can be reasonably expected to cause the failure of the life support device or system, or to affect its safety or effectiveness.

### **BANNED SUBSTANCE COMPLIANCE**

National Semiconductor certifies that the products and packing materials meet the provisions of the Customer Products Stewardship Specification (CSP-9-111C2) and the Banned Substances and Materials of Interest Specification (CSP-9-111S2) and contain no "Banned Substances" as defined in CSP-9-111S2.



National Semiconductor Americas Customer Support Center

Email: new.feedback@nsc.com Tel: 1-800-272-9959

www.national.com

National Semiconductor Europe Customer Support Center Fax: +49 (0) 180-530 85 86

Email: europe.support@nsc.com
Deutsch Tel: +49 (0) 69 9508 6208
English Tel: +44 (0) 870 24 0 2171
Français Tel: +33 (0) 1 41 91 8790

National Semiconductor
Asia Pacific Customer
Support Center
Email: ap.support@nsc.com

National Semiconductor Japan Customer Support Center Fax: 81-3-5639-7507 Email: jpn.feedback@nsc.com Tel: 81-3-5639-7560