ISD5008

3V, SINGLE-CHIP

VOICE RECORD/PLAYBACK DEVICE

4- TO 8-MINUTES DURATION

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1 GENERAL DESCRIPTION

The ISD5008 ChipCorder product is a 3V fully-integrated, single-chip solution which provides seamless integration of enhanced voice record and playback features for digital cellular phones (GSM, CDMA, TDMA, PDC, and PHS), automotive communications, GPS/navigation systems, and portable communication products. This low-power, 3-volt device enables customers to quickly and easily integrate 4 to 8 minutes of voice storage features such as one-way or two-way (full duplex) call record, voice memo record, and call screening/answering machine functionality.

Like other ChipCorder products, the ISD5008 integrates the sampling clock, anti-aliasing and smoothing filters, and the Multi-Level Storage (MLS) array into a single chip. For enhanced voice features, the ISD5008 eliminates external circuitry mostly by also integrating automatic gain control (AGC), a power amplifier/speaker driver, volume control, summing amplifiers, analog switches, and a car kit interface. Input level adjustable amplifiers are also implemented, providing a flexible interface for multiple applications.

Duration/sample rate selection is accomplished via software, allowing customers to optimize quality and duration for various features within the same end product.

The ISD5008 device is designed for use in a microprocessor- or microcontroller-based system. Address, control, and duration selection are accomplished through a Serial Peripheral Interface (SPI) or Microwire Serial Interface to minimize pin count.

Recordings are stored into on-chip non-volatile memory cells, providing zero-power message storage. This unique, single-chip solution is made possible through Nuvoton's patented MLS technology. Voice and audio signals are stored directly into solid-state memory in their natural, uncompressed form, providing superior quality voice and music reproduction.

2 FEATURES

Fully-Integrated Solution

- Single-chip voice record/playback solution
- Integrated sampling clock, anti-aliasing and smoothing filters, and MLS array
- Integrated analog features such as automatic gain control (AGC), audio gating switches, speaker driver, summing amplifiers, volume control, and AUX IN/AUX OUT interface (e.g., for car kits)

Low-Power Consumption

- Single +3 volt supply
- Operating current:
  - I_{CC_Play} = 15 mA (typical)
  - I_{CC_Rec} = 25 mA (typical)
  - $I_{CC_Feedthru}$ = 12 mA (typical)
- Standby current:
  - I_{SB} = 1 µA (typical)
- Power consumption controlled by SPI or Microwire control register
- Most stages can be individually powered down for minimum power consumption

Enhanced Voice Features

- One or two-way (full duplex) conversation record (record signal summation)
- One- or two-way (full duplex) message playback (while on a call)
- Voice memo record and playback
- Private call screening
- Answering machine
- Personalized outgoing message (given caller ID information from host chip set)
- Private call announce while on call (given CIDCW information from host chip set)

Easy-to-Use and Control

- No compression algorithm development required
- User-selectable sampling rates of 8.0 kHz, 6.4 kHz, 5.3 kHz, or 4.0 kHz
- Microcontroller SPI or Microwire™ Serial Interface
- Fully addressable to handle multiple messages

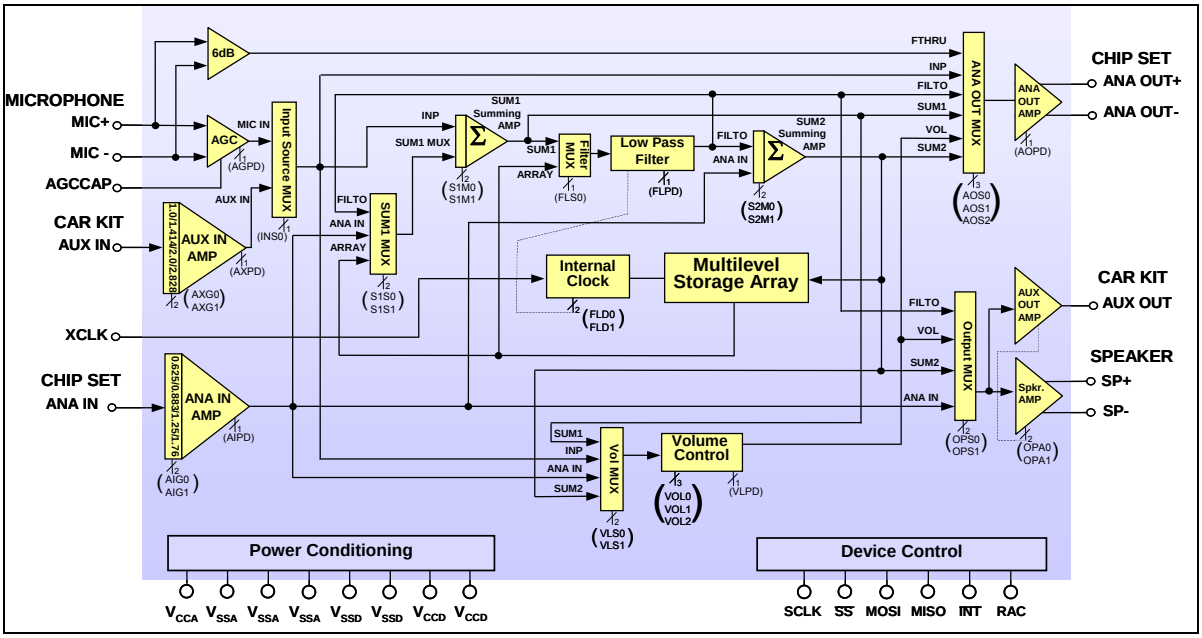
High Quality Solution

- High quality voice and music reproduction
- Standard 100-year message retention (typical)
- 100,000 record cycles (typical)

Options

- Available in die, PDIP, SOIC, TSOP, and chip scale packaging (CSP)
- Compact µBGA chip scale package available for portable applications
- Temperature : Commercial, Extended and Industrial

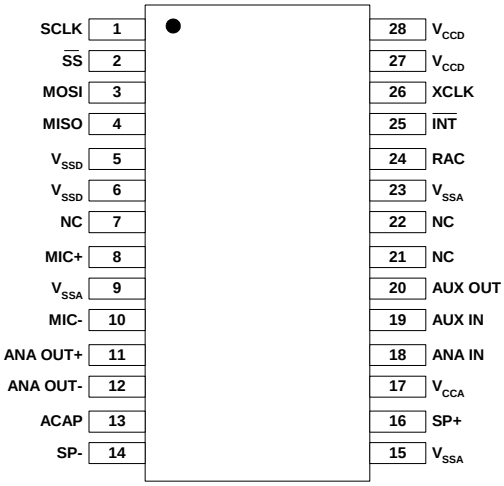
3 BLOCK DIAGRAM



4 PIN CONFIGURATION



TSOP



PDIP/SOIC

5 PIN DESCRIPTION

PIN NAME	PDIP/SOIC	TSOP	FUNCTION
SCLK	1	8	**Serial Clock**: The SCLK is the clock input to the ISD5008. Generated by the master microcontroller, the SCLK synchronizes data transfers in and out of the device through the MISO and MOSI lines.
$\overline{SS}$	2	9	**Slave Select**: This input, when LOW, selects the device.
MOSI	3	10	**Master Out Slave In**: MOSI is the serial data input to the ISD5008 device. The master microcontroller places data to be clocked into the ISD5008 device on the MOSI line one-half cycle before the rising edge of SCLK. Data is clocked into the device with LSB (Least Significant Bit) first.
MISO	4	11	**Master In Slave Out**: MISO is the serial data output of the ISD5008. Data is clocked out on the falling edge of SCLK. This output goes into a high-impedance state when the device is not selected. Data is clocked out of the device with LSB first.
V_{SSD} / V_{SSA}	5, 6 / 9, 15, 23	12, 13 / 2, 15, 22	**Ground**: The ISD5008 utilizes separate analog and digital ground busses. The analog ground (V_{SSA}) pins should be tied together as close to the package as possible and connected through a low-impedance path to power supply ground. The digital ground (V_{SSD}) pin should be connected through a separate low-impedance path to power supply ground. These ground paths should be large enough to ensure that the impedance between the V_{SSA} pins and the V_{SSD} pins is less than 3 Ω. The backside of the die is connected to V_{SSD} through the substrate resistance. In a chip-on-board design, the die attach area must be connected to V_{SSD}.
MIC+ / MIC-	8 / 10	16 / 17	**Microphone +/–**: The microphone inputs transfer the voice signal to the on-chip AGC preamplifier or directly to the ANA OUT MUX, depending on the selected path. The AGC circuit has a range of 45dB in order to deliver a nominal 694 mVp-p into the storage array from a typical electret microphone output of 2 to 20 mVp-p. The direct path to the ANA OUT MUX has a gain of 6dB so a 208 mVp-p signal across the differential microphone inputs would give 416 mVp-p across the ANA OUT pins. The input impedance is typically 10 kΩ.
ANAOUT+ / ANAOUT-	11 / 12	18 / 19	**Analog Outputs**: These differential outputs are designed to match to the microphone input of the telephone chip set. It is designed to drive a minimum of 5 kΩ between the "+" and "–" pins to a nominal voltage level of 700 mVp-p. Both pins have DC bias of approximately 1.2 VDC. The AC signal is superimposed upon this analog ground voltage. These pins can be used single-ended, getting only half the voltage. Do **NOT** ground the unused pin.

PIN NAME	PDIP/SOIC	TSOP	FUNCTION
ACAP	13	20	**AGC Capacitor**: This pin provides the capacitor connection for setting the parameters of the microphone AGC circuit. It should have a 4.7 µF capacitor connected to ground. **It cannot be floating.** This is because the capacitor is also used in the playback mode for the AutoMute circuit. This circuit reduces the amount of noise present in the output during quiet pauses. Tying this pin to ground gives maximum gain, or to V_{CCA} gives minimum gain for the AGC amplifier but will cancel the AutoMute function.
SP- / SP+	14 / 16	21 / 23	**Speaker Outputs**: This differential speaker output is designed to drive an 8Ω speaker up to a maximum power of 23.5 mW. This stage has two selectable gains, 1.32 and 1.6, which can be chosen through the configuration registers. These pins are biased to approximately 1.2 VDC and, if used single-ended, must be capacitively coupled to their load. Do **NOT** ground the unused pin.
V_{CCA} / V_{CCD}	17 / 27, 28	24 / 6, 7	**Power Supplies**: To minimize noise, the analog and digital circuits in the ISD5008 device uses separate power busses. These +3V busses lead to separate pins. Tie the V_{CCD} pins together as close as possible and decouple both supplies as near to the package as possible
ANA IN	18	25	**Analog Input**: The ANA IN pin is the analog input from the telephone chip set. It can be switched (by the SPI bus) to the speaker output, the array input or to various paths. This pin is designed to accept a nominal 1.11 Vp-p when at its minimum gain (6dB) setting. See Table 4. There is additional gain available in 3dB steps controlled from the SPI bus, if required, up to 15dB.
AUX IN	19	26	**Auxiliary Input**: The AUX IN is an additional audio input to the ISD5008, such as from the microphone circuit in a mobile phone "car kit." This input has a nominal 700 mVp-p level at its minimum gain setting (0dB). See Table 5. Additional gain is available in 3 dB steps (controlled by the SPI bus) up to 9dB.
AUX OUT	20	27	**Auxiliary Output**: The AUXOUT is an additional audio output pin, to be used, for example, to drive the speaker circuit in a "car kit." It drives a minimum load of 5 kΩ and up to a maximum of 1 Vp-p. The AC signal is superimposed on approximately 1.2 VDC bias and must be capacitively coupled to the load.

PIN NAME	PDIP/SOIC	TSOP	FUNCTION
RAC	24	3	**Row Address Clock**: RAC is an open drain output pin that marks the end of a row. At the 8 kHz sampling frequency, the duration of this period is 200 ms. There are 1,200 rows of memory in the ISD5008 devices. RAC stays HIGH for 175 ms and stays LOW for the remaining 25 ms before it reaches the end of the row. At the 8 kHz sampling frequency, the RAC pin remains HIGH for 109.38 µsec and stays LOW for 15.63 µsec under the Message Cueing mode. See Table 15 Timing Parameters for RAC timing information at other sample rates. When a record command is first initiated, the RAC pin remains HIGH for an extra T_{RACLO} period, to load sample and hold circuits internal to the device. The RAC pin can be used for message management techniques. A pull-up resistor is required to connect this pin to other device.
$\overline{INT}$	25	4	**Interrupt**: $\overline{INT}$ is an open drain output pin. The ISD5008 interrupt pin goes LOW and stays LOW when an Overflow (OVF), or End of Message (EOM) marker or Message Cueing is detected. The interrupt is cleared the next time an SPI cycle is completed. The interrupt status can be read by a RINT instruction that will give one of the two flags out the MISO line. A pull-up resistor is required to connect this pin to other device. ***OVF Flag***. The overflow flag indicates that the end of the ISD5008's analog memory has been reached during a record or playback operation. ***EOM Flag.*** The end of message flag is set only during playback, when an EOM is found. There are eight possible EOM markers per row.
XCLK	26	5	**External Clock**: The external clock input has an internal pull-down device. Normally, the ISD5008 is operated at one of four internal rates selected for its internal oscillator by the Sample Rate Select bits. If greater precision is required, the device can be clocked through the XCLK pin as described in Table 2. Because the anti-alising and smoothing filters track the Sample Rate Select bits, one must, for optimum performance, change the external clock AND the Sample Rate Configuration bits to one of the four values properly to set the filters to the correct cutoff frequency as described in Table 1. The duty cycle on the input clock is not critical, as the clock is immediately divided by two internally. If the XCLK is not used, this input should be connected to V_{SSD}.

6 FUNCTIONAL DESCRIPTION

6.1 DETAILED DESCRIPTION

6.1.1 Speech/Sound Quality

The ISD5008 ChipCorder product can be configured via software to operate at 4.0, 5.3, 6.4, or 8.0 kHz sampling frequency, allowing the user a choice of speech quality options. Increasing the duration decreases the sampling frequency and bandwidth, which affects sound quality. Table 1 shows the relationship between sampling frequency, duration and filter pass band.

The speech samples are stored directly into on-chip non-volatile memory without the digitization and compression associated with other solutions. Direct analog storage provides a natural sounding reproduction of voice, music, tones, and sound effects not available with most solid-state solutions.

6.1.2 Duration

To meet end system requirements, the ISD5008 device is a single-chip solution which provides from 4 to 8 minutes of voice record and playback, depending on the sample rates defined by customer software.

TABLE 1: SAMPLING RATE / DURATION / FILTER EDGE

Sample Rate (kHz)	Duration (Minutes)	Filter Pass Band* (kHz)
8	4.0	3.4
6.4	5.0	2.7
5.3	6.0	2.3
4	8.0	1.7

* -3dB point

6.1.3 Flash Storage

One of the benefits of Nuvoton's ChipCorder technology is the use of on-chip nonvolatile memory, which provides zero- power message storage. The message is retained for up to 100 years (typically) without power. In addition, the device can be re-recorded over 100,000 times (typically).

6.1.4 Microcontroller Interface

A four-wire (SCLK, MOSI, MISO, SS) SPI interface is provided for ISD5008 control, addressing functions, and sample rate selection. The ISD5008 is configured to operate as a peripheral slave device with a microcontroller-based SPI bus interface. Read/Write access to all the internal registers occurs through this SPI interface. An interrupt signal (INT) and internal read-only Status Register are provided for handshake purposes.

6.1.5 Memory Architecture

The ISD5008 device contains a total of 1,920K Flash memory cells, which is organized as 1,200 rows of 1,600 cells each. The duration is counted according to the number of rows, while the row number is represented by the related 16 address bits of MOSI as described in the SPI section.

6.1.6 Programming

The ISD5008 device is also ideal for playback-only applications, where single- or multiple-message playback is controlled through the SPI port. Once the desired message configuration is created, duplicates can easily be generated via a third-party programmer. For more information on available application tools and programmers, please see the Nuvoton website at www.Nuvoton-usa.com.

TABLE 2: EXTERNAL CLOCK INPUT

Duration (Minutes)	Sample Rate (kHz)	Required Clock (kHz)
4	8.0	1024
5	6.4	819.2
6	5.3	682.7
8	4.0	512

TABLE 3: INTERNAL SAMPLING RATE / FILTER EDGE

FLD1	FLD0	Sample Rate (kHz)	Filter Pass Band (kHz)
0	0	8	3.4
0	1	6.4	2.7
1	0	5.3	2.3
1	1	4	1.7

6.2 ANALOG FUNCTIONAL PINS

6.2.1 Mic+, Mic-

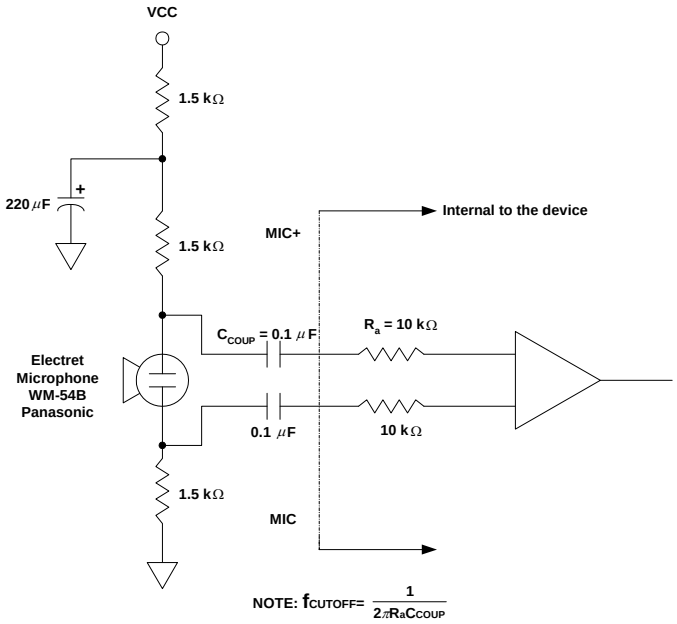
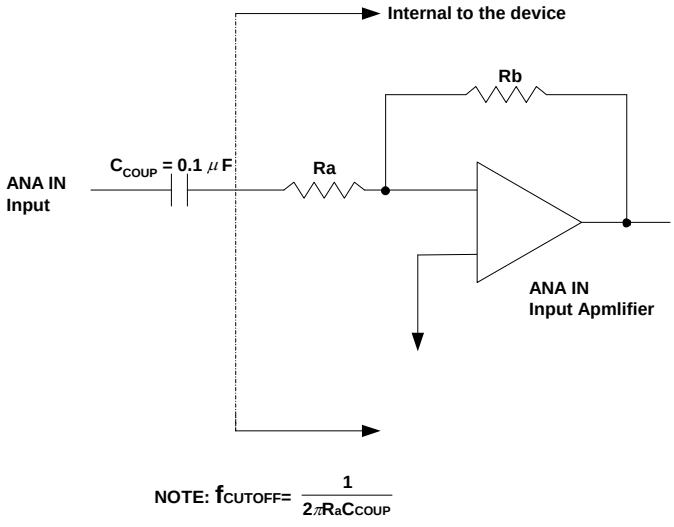


FIGURE 1: MICROPHONE INPUT

6.2.2 ANA IN (Analog Input)



Gain Setting	Resistor Ratio (Rb/Ra) kΩ/kΩ	Gain	Gain [2] (dB)
00	63.9/102	0.625	-4.1
01	77.9/88.1	0.88	-1.1
10	92.3/73.8	1.25	1.9
11	106/60	1.77	4.9

FIGURE 2: ANA IN INPUT MODES

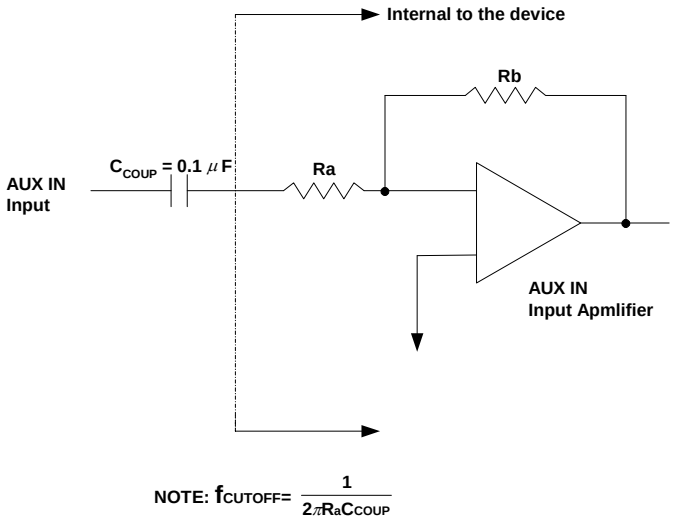
TABLE 4: ANA IN AMPLIFIER GAIN SETTINGS

Setting (1)	0TLP Input V_{PP} (3)	CFG0		Gain (2)	Array In/Out V_{PP}	Speaker Out V_{PP} (4)
		AIG1	AIG0			
6 dB	1.11	0	0	.625	.694	2.22
9 dB	.785	0	1	.883	.694	2.22
12 dB	.555	1	0	1.250	.694	2.22
15 dB	.393	1	1	1.767	.694	2.22

NOTES:

1. Gain from ANA IN to SP+ / _
2. Gain from ANA In to ARRAY IN
3. 0TLP Input is the reference Transmission Level Point that is used for testing. This level is typically 3 dB below clipping.
4. Speaker Out gain set to 1.6 (High). (Differential)

6.2.3 AUX IN (Auxillary Input)



Gain Setting	Resistor Ratio (Rb/Ra) kΩ/kΩ	Gain	Gain [2] (dB)
00	40.1/40.1	1	0
01	47.0/33.2	1.414	3
10	53.5/26.7	2	6
11	58.2/21.0	2.82	9

FIGURE 3: AUX IN INPUT MODES

TABLE 5: AUXIN AMPLIFIER GAIN SETTINGS

Setting [1]	0TLP Input V_{PP} [3]	CFG0		Gain [2]	Array In/Out V_{PP}	ANA OUT V_{PP} [4]
		AXG1	AXG0			
0 dB	.694	0	0	1.00	.694	.694
3 dB	.491	0	1	1.41	.694	.694
6 dB	.347	1	0	2.00	.694	.694
9 dB	.245	1	1	2.82	.694	.694

NOTES:

1. Gain from AUX IN to ANA OUT
2. Gain from AUX IN to ARRAY IN
3. 0TLP Input is the reference Transmission Level Point that is used for testing. This level is typically 3 dB below clipping.
4. Differential

6.2.4 ACAP (AGC Capacitor)

This pin provides the capacitor connection for setting the parameters of the microphone AGC circuit. It should have a 4.7 μF capacitor connected to ground. It cannot be left floating. This is because the capacitor is also used for the AutoMute circuit. This circuit reduces the amount of noises present in the output during quiet pauses. Tying this pin to ground gives maximum gain; to V_{CCA} gives minimum gain for the AGC amplifier but will cancel the AutoMute function.

6.3 INTERNAL FUNCTIONAL BLOCKS

FIGURE 7: MICROPHONE AMPLIFIER

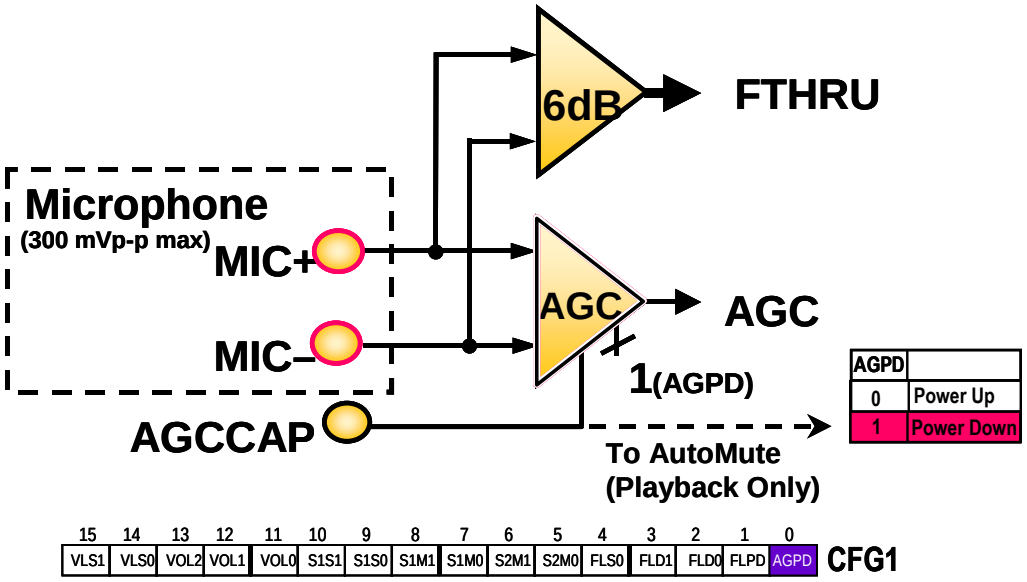


FIGURE 7: ANA IN and AUX IN

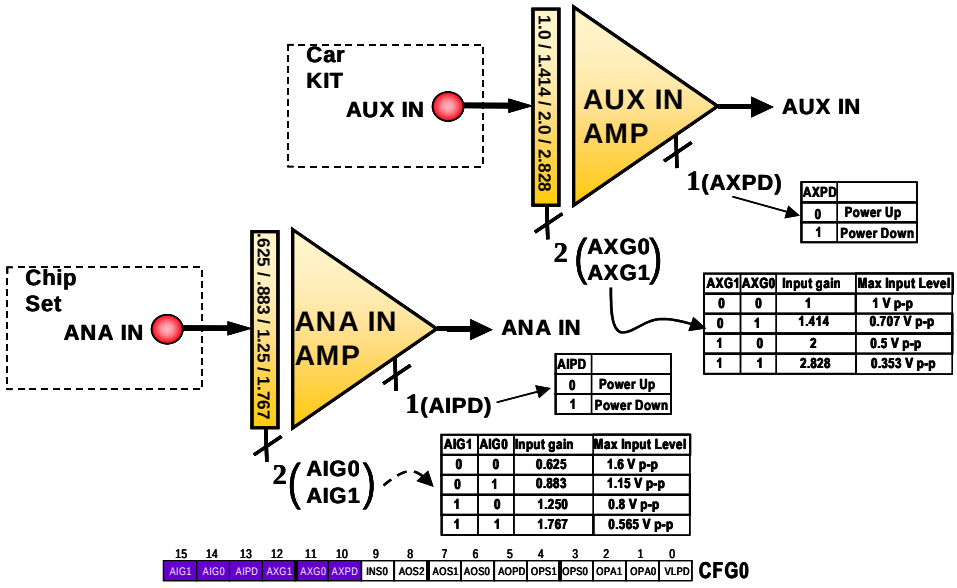


FIGURE 7: ISD5008 CORE (LEFT HALF)

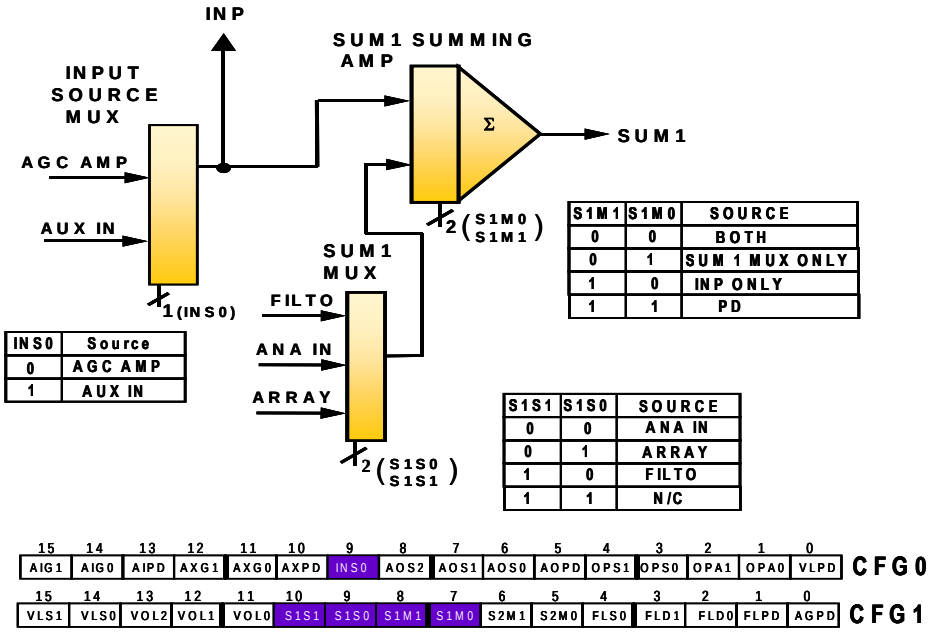


FIGURE 8: ISD5008 CORE (RIGHT HALF)

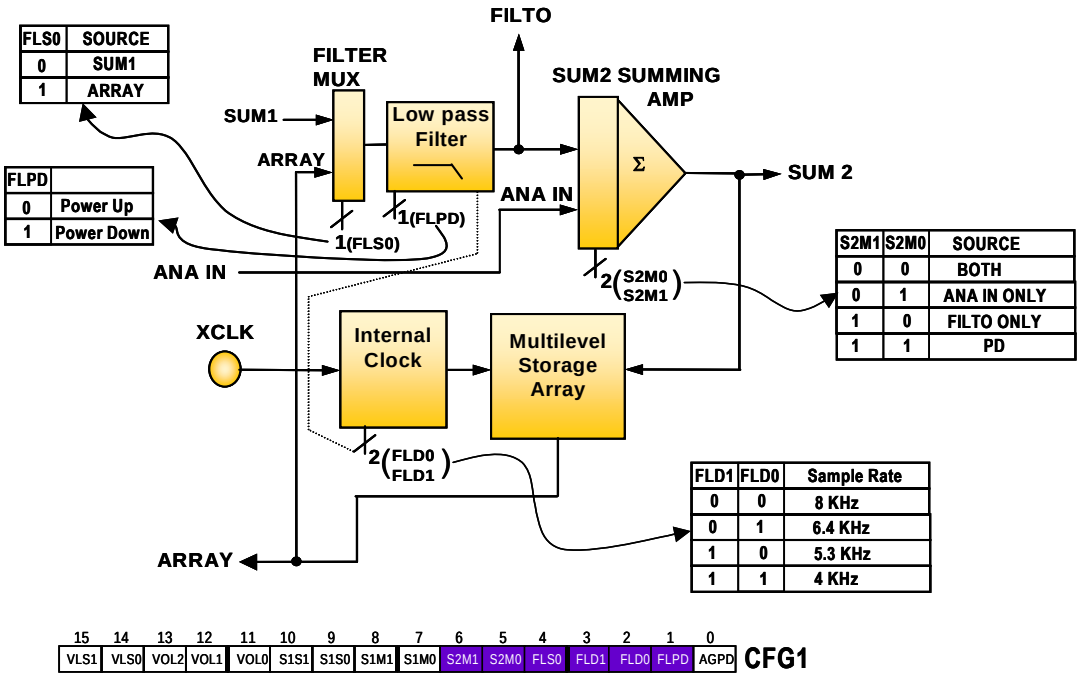
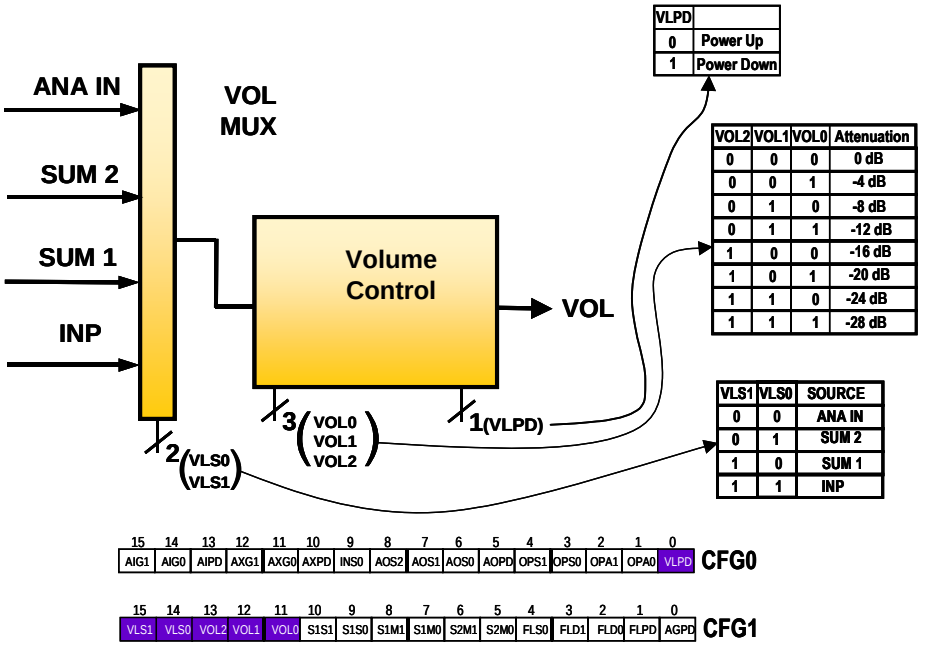


FIGURE 9: VOLUME CONTROL



VLPD	
0	Power Up
1	Power Down

VOL2	VOL1	VOL0	Attenuation
0	0	0	0 dB
0	0	1	-4 dB
0	1	0	-8 dB
0	1	1	-12 dB
1	0	0	-16 dB
1	0	1	-20 dB
1	1	0	-24 dB
1	1	1	-28 dB

VLS1	VLS0	SOURCE
0	0	ANA IN
0	1	SUM 2
1	0	SUM 1
1	1	INP

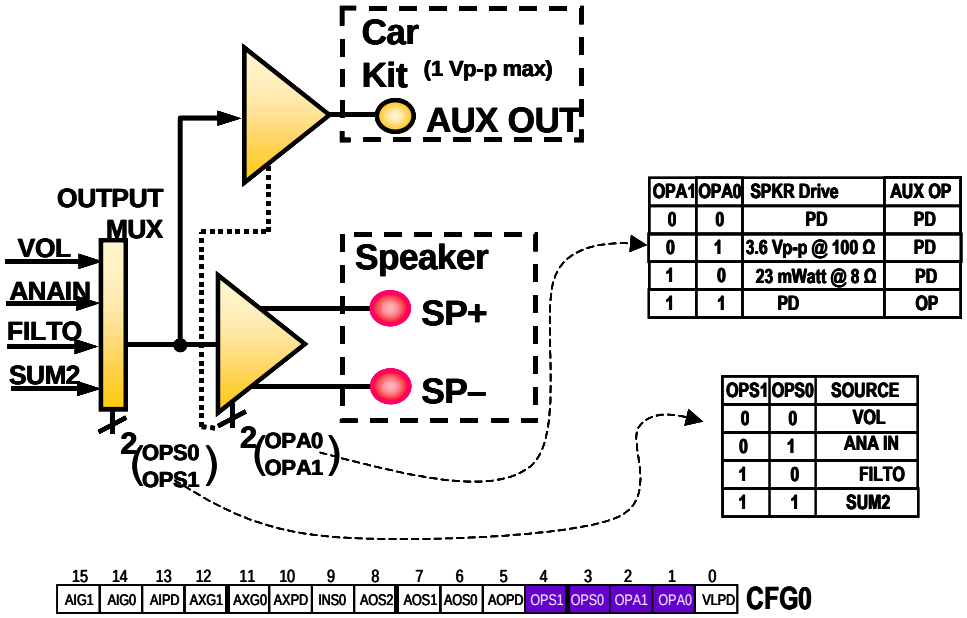
15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0
AIG1	AIG0	AIPD	AXG1	AXG0	AXPD	INS0	AOS2	AOS1	AOS0	AOPD	OPS1	OPS0	OPA1	OPA0	VLPD

CFG0

15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0
VLS1	VLS0	VOL2	VOL1	VOL0	S1S1	S1S0	S1M1	S1M0	S2M1	S2M0	FLS0	FLD1	FLD0	FLPD	AGPD

CFG1

FIGURE 10: SPEAKER and AUX OUT



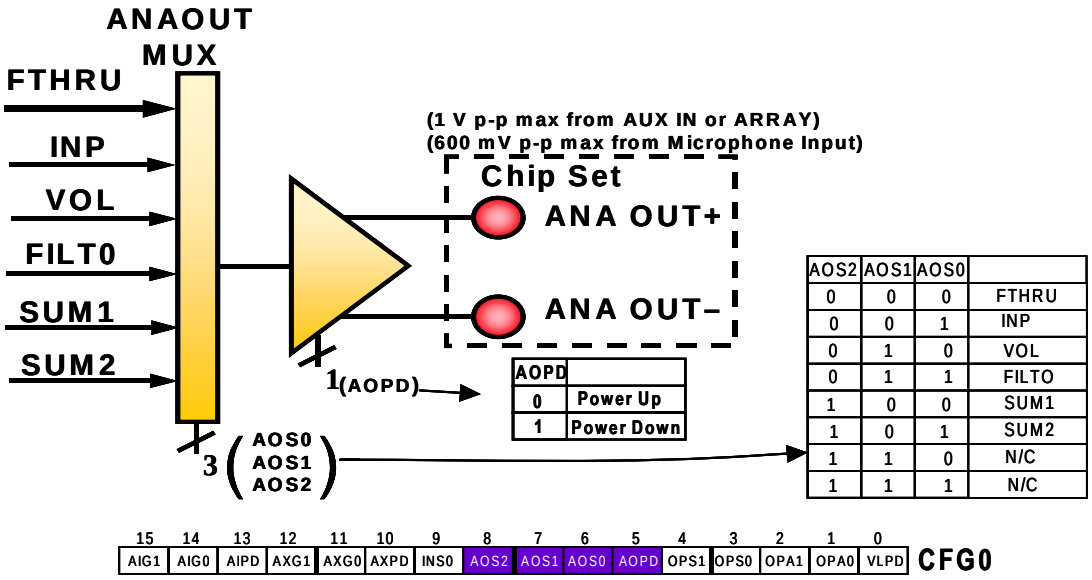
OPA1	OPA0	SPKR Drive	AUX OP
0	0	PD	PD
0	1	3.6 Vp-p @ 100 Ω	PD
1	0	23 mWatt @ 8 Ω	PD
1	1	PD	OP

OPS1	OPS0	SOURCE
0	0	VOL
0	1	ANA IN
1	0	FILTO
1	1	SUM2

15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0
AIG1	AIG0	AIPD	AXG1	AXG0	AXPD	INS0	AOS2	AOS1	AOS0	AOPD	OPS1	OPS0	OPA1	OPA0	VLPD

CFG0

FIGURE 11: ANA OUT Output



6.4 SERIAL PERIPHERAL INTERFACE (SPI) DESCRIPTION

The ISD5008 product operates from a SPI serial interface, which operates with the following protocol:

The data transfer protocol assumes that the microcontroller's SPI shift registers are clocked on the falling edge of the SCLK. However, for the ISD5008, data is clocked into the MOSI pin at the rising clock edge, while data is clocked out onto the MISO pin at the falling clock edge.

1. All serial data transfers begin with the falling edge of SS pin.
2. SS is held LOW during all serial communications and held HIGH between instructions.
3. Data is clocked in on the rising clock edge and data is clocked out on the falling clock edge.
4. Play and Record operations are initiated by enabling the device by asserting the SS pin LOW, shifting in an opcode and an address field to the ISD5008 device (refer to the Opcode Summary of Table 6).
5. The opcodes and address fields are as follows: <8 control bits> and <16 address bits>.
6. Each operation that ends in an EOM or Overflow will generate an interrupt, including the Message Cueing cycles. The Interrupt will be cleared the next time an SPI cycle is completed.
7. As Interrupt data is shifted out of the ISD5008 MISO pin, control and address data is simultaneously being shifted into the MOSI pin. Care should be taken such that the data shifted in is compatible with current system operation. It is possible to read interrupt data and start a new operation within the same SPI cycle.
8. A record or playback operation begins with the RUN bit set and the operation ends with the RUN bit reset.
9. All operations begin with the rising edge of SS.

6.4.1 Message Cuing

Message cueing allows the user to skip through messages, without knowing the actual physical location of the message. This operation is used during playback. In this mode, the messages are skipped 1600 times faster than in normal playback mode. It will stop when an EOM marker is reached. Then, the internal address counter will point to the next message.

6.4.2 Opcodes

TABLE 6: OPCODE SUMMARY

Instruction	Opcode <8 bits> [1] Address <16 bits>	Operational Summary
POWERUP	0110 0000	Power-Up: Power-Up the device
LOADCFG0 [2]	01X0 0010 <D15-D0>	Loads a 16-bit value into Configuration Register 0
LOADCFG1	01X0 0100 <D15-D0>	Loads a 16-bit value into Configuration Register 1
SETPLAY	1110 0000 <A15-A0>	Initiates Playback from address <A15-A0>
PLAY	1111 0000	Playback from current address (until EOM or OVF)
SETREC	1010 0000 <A15-A0>	Initiates Record at address <A15-A0>
REC	1011 0000	Records from current address until OVF is reached
SETMC	1110 1000 <A15-A0>	Initiates Message Cueing (MC) from address <A15-A0>
MC	1111 1000	Performs a Message Cue. Proceeds to the end of the current message (EOM) or enters OVF condition if it reaches the end of the array.
STOP	0111 0000	Stops current operation
STOPWRDN	0101 0000	Stops current operation and enters stand-by (power-down) mode.
RINT	0111 0000	Read interrupt status bits: OVF and EOM.

NOTES:

[1] X = Don't Care.

[2] Changes in CFG0 are not recognized until CFG1 is loaded. The changes will occur at the rising edge of $\overline{SS}$ during the cycle that CFG1 is loaded.

6.4.3 Power-Up Sequence

The ISD5008 will be ready for an operation after T_{PUD} (25 ms approximately for 8 kHz sample rate). The user needs to wait T_{PUD} before issuing an instruction. Below are suggested playback and record examples for references.

6.4.3.1 Record Mode

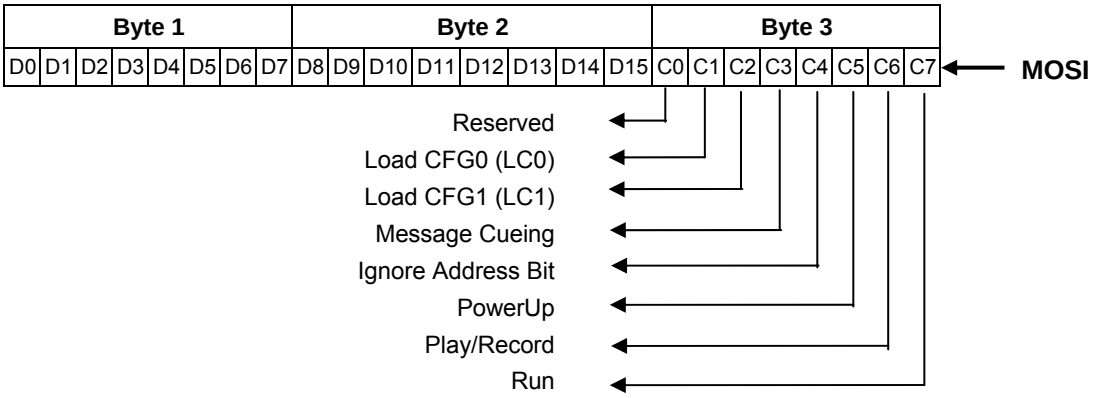
1. Send POWERUP command.
2. Wait T_{PUD} (power-up delay).
3. Send POWERUP command.
4. Wait 2 x T_{PUD} (power-up delay).
5. Load CFG0 and CFG1 for desired operation.
6. Wait $T_{PUD.}$
7. Send SETREC command with address xx, or send REC command.
8. Send STOP to halt the record operation or when the end of memory (OVF) is reached, then record stops automatically.
9. Wait $T_{Stop/Pause.}$.

6.4.3.2 Playback Mode

1. Send POWERUP command.
2. Wait T_{PUD} (power-up delay).
3. Load CFG0 and CFG1 for desired operation.
4. Wait $T_{PUD.}$
5. Send SETPLAY command with address xx, or send PLAY command.
6. Send STOP to halt the playback operation or wait until an EOM is reached, then playback stops automatically.
7. Wait $T_{Stop/Pause.}$.

6.4.4 SPI Port

The following diagram describes the SPI port and the control bits associated with it.



6.4.5 SPI Control Register

The SPI control register provides control of individual device functions such as Play, Record, Message Cueing, Power-Up and Power-Down, Start and Stop operations, Ignore Address Pointers and Load Configuration Registers.

TABLE 7: SPI CONTROL REGISTER

Control Register	Bit	Device Function
RUN		Enable or Disable an operation
=	1	Start
=	0	Stop
$P/\overline{R}$		Selects Play or Record operation
=	1	Play
=	0	Record
MC		Enable or Disable Message Cueing
=	1	Enable Message Cueing
=	0	Disable Message Cueing
LC0		
=	1	Load Configuration Reg 0
=	0	No Load

Control Register	Bit	Device Function
PU		Master power control
=	1	Power-Up
=	0	Power-Down
IAB		Ignore address control bit
=	1	Ignore input address register (A15-A0)
=	0	Use the input address register contents for an operation (A15-A0)
A15-A0		Output of the row pointer register
D15-D0		Input control and address register
LC1		
=	1	Load Configuration Reg 1
=	0	No Load

TABLE 8: CONFIGURATION REGISTER 0

D15	D14	D13	D12	D11	D10	D9	D8	D7	D6	D5	D4	D3	D2	D1	D0	CFGO
AIG1	AIG0	AIPD	AXG1	AXG0	AXPD	INS0	AOS2	AOS1	AOS0	AOPD	OPS1	OPS0	OPA1	OPA0	VLPD	
ANA IN AMP Gain SET (2 bits)		ANA IN Power Down	AUX IN AMP Gain SET (2 bits)		AUX IN Power Down	INPUT SOURCE MUX Select (1 bit)	ANA OUT MUX Select (3 bits)			ANA OUT Power Down	OUTPUT MUX Select (2 bits)		SPKR & AUX OUT Control (2 bits)		Volume Control Power Down	

NOTE: See details on following pages

TABLE 9: CONFIGURATION REGISTER 1

D15	D14	D13	D12	D11	D10	D9	D8	D7	D6	D5	D4	D3	D2	D1	D0	CFG1
VLS1	VLS0	VOL2	VOL1	VOL0	S1S1	S1S0	S1M1	S1M0	S2M1	S2M0	FLSO	FLD1	FLD0	FLPD	AGPD	
VOLUME CONT. MUX Select (2 bits)		VOLUME CONTROL (3 bits)			SUM 1 MUX Select (2 bits)		SUM 1 SUMMING AMP Control (2 bits)		SUM 2 SUMMING AMP Control (2 bits)		FILTER MUX Select	SAMPLE RATE (& Filter) Set Up (2 bits)		Filter Power Down	AGC AMP Power Down	

NOTE: See details on following pages

Detail of Configuration Register 0		
Volume Control Power Bit	Bit 0 (VLPD)	0 = Power ON 1 = Power OFF
SPEAKER and AUX OUT Control Bits	Bits 2,1 (OPA1, OPA0)	00 = Power down SPKR and AUX 01 = SPKR ON, HIGH GAIN, AUX Power down 10 = SPKR ON, LOW GAIN, AUX Power down 11 = SPKR Powered down, AUX ON
OUTPUT MUX Con trol Bits	Bits 4,3 (OPS1, OPS0)	00 = Source is VOL CONTROL (VOL) 01 = Source is ANA IN Input (ANA IN AMP) 10 = Source is LOW PASS FILTER (FILT0) 11 = Source is SUM2 SUMMING AMP (SUM2)
ANA OUT Power Bit	Bit 5 (AOPD)	0 = Power ON 1 = Power OFF
ANA OUT MUX Con trol Bits	Bits 8,7,6 (AOS2, AOS1, AOS0)	000 = Source is MICROPHONE AMP (FTHRU) 001 = Source is INPUT MUX (INP) 010 = Source is VOLUME CONTROL (VOL) 011 = Source is LOW PASS FILTER (FILT0) 100 = Source is SUM1 SUMMING AMP (SUM1) 101 = Source is SUM2 SUMMING AMP (SUM2) 110 = Unused 111 = Unused
INPUT SOURCE MUX Control Bit	Bit 9 (INS0)	0 = Source is Microphone AGC AMP (AGC) 1 = Source is AUX IN Input (AUX IN AMP)
AUX IN AMP Power Bit	Bit 10 (AXPD)	0 = Power ON 1 = Power OFF
AUX IN AMP Control Bits	Bits 12,11 (AXG1, AXG0)	00 = Input Gain = 1, OTLP input Level = 0.694 01 = Input Gain = 1.414, OTLP input Level = 0.491 10 = Input Gain = 2, OTLP input Level = 0.347 11 = Input Gain = 2.828, OTLP input Level = 0.245
ANA IN AMP Power Bit	Bit 13 (AIPD)	0 = Power ON 1 = Power OFF
ANA IN AMP Control Bits	Bits 15,14 (AIG1, AIG0)	00 = Input Gain = 0.625, OTLP input Level = 1.11 01 = Input Gain = 0.883, OTLP input Level = 0.7I85 10 = Input Gain = 1.250, OTLP input Level = 0.555 11 = Input Gain = 1.767, OTLP input Level = 0.393

Detail of Configuration Register 1		
AGC Power Control Bit	Bit 0 (AGPD)	0 = Power ON 1 = Power OFF
LOW PASS FILTER Power Control Bit	Bit 1 (FLPD)	0 = Power ON 1 = Power OFF
SAMPLE RATE and LOW PASS FILTER Control Bits	Bits 3,2 (FLD1, FLD0)	00 = Sample Rate = 8 KHz, FPB = 3.4 KHz 01 = Sample Rate = 6.4 KHz, FPB = 2.7 KHz 10 = Sample Rate = 5.3 KHz, FPB = 2.3 KHz 11 = Sample Rate = 4 KHz, FPB = 1.7 KHz
FILTER MUX Control bits	Bit 4 (FLS0)	0 = Source is SUM1 SUMMING AMP (SUM1) 1 = Source is Analog Memory Array (ARRAY)
SUM 2 SUMMING AMP Control Bits	Bits 6,5 (S2M1, S2M0)	00 = Source is both ANA IN AMP and FILT0 01 = Source is ANA IN Input (ANA IN AMP) ONLY 10 = Source is LOW PASS FILTER (FILT0) ONLY 11 = Power Down SUM2 SUMMING AMP
SUM1 SUMMING AMP Control Bits	Bit 8,7 (S1M1, S1M0)	00 = Source is both SUM1 and INP 01 = Source is SUM1 SUMMING AMP (SUM1) ONLY 10 = Source is INPUT MUX (INP) ONLY 11 = Power Down SUM1 SUMMING AMP
SUM1MUX Control Bits	Bit 10,9 (S1S1, S1S0)	00 = Source is ANA IN Input (ANA IN AMP) 01 = Source is Analog Memory Array (ARRAY) 10 = Source is LOW PASS FILTER (FILT0) 11 = UNUSED
VOLUME CONTROL Control Bits	Bits 13,12,11 (VOL2, VOL1, VOL0)	000 = Attenuation = 0 dB 001 = Attenuation = 4 dB 010 = Attenuation = 8 dB 011 = Attenuation = 12 dB 100 = Attenuation = 16 dB 101 = Attenuation = 20 dB 110 = Attenuation = 24 dB 111 = Attenuation = 28 dB
VOL MUX Control Bits	Bit 15,14 (VLS1, VLS0)	00 = Source is ANA IN Input (ANA IN AMP) 01 = Source is SUM2 SUMMING AMP (SUM2) 10 = Source is SUM1 SUMMING AMP (SUM1) 11 = Source is INPUT MUX (INP)

Configuration Register Notes

1. ***Important***: All changes to the internal settings of the ISD5008 are synchronized with the load of Configuration Register 1. A command to load Configuration Register 1 immediately transfers the input data to the internal settings of the device and the changes take place immediately at the end of the command when $\overline{SS}$ goes HIGH. A load to Configuration Register 0 sends the new data to

a temporary register in the ISD5008 and does not affect the internal settings of the device. The next time Configuration Register 1 is loaded, data will also transfer from the temporary register to the Configuration Register 0 and effect the desire changes. See Figure & Table 13.

2. Configuration Registers may be loaded with data at any time, including when the chip is powered down using the PU bit in the SPI Control Resgister. The PU bit in the SPI Control Word will have to be set to a "1" before the changes in configurarion will be seen.

FIGURE 13: Configuration Register Programming Sequence

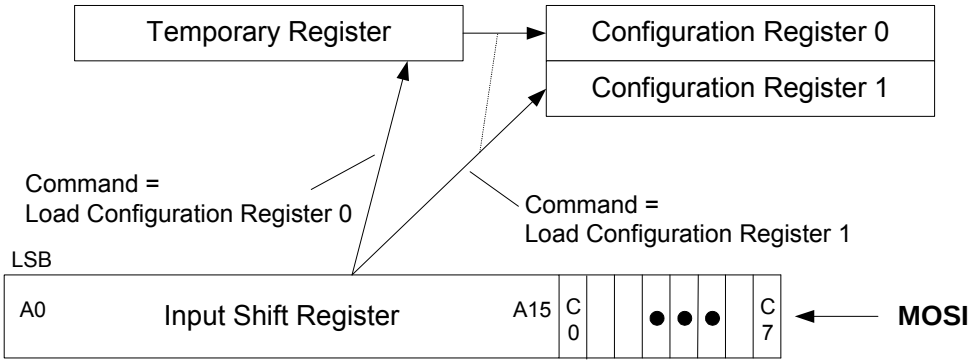


FIGURE 14: SPI Interface Simplified Block Diagram

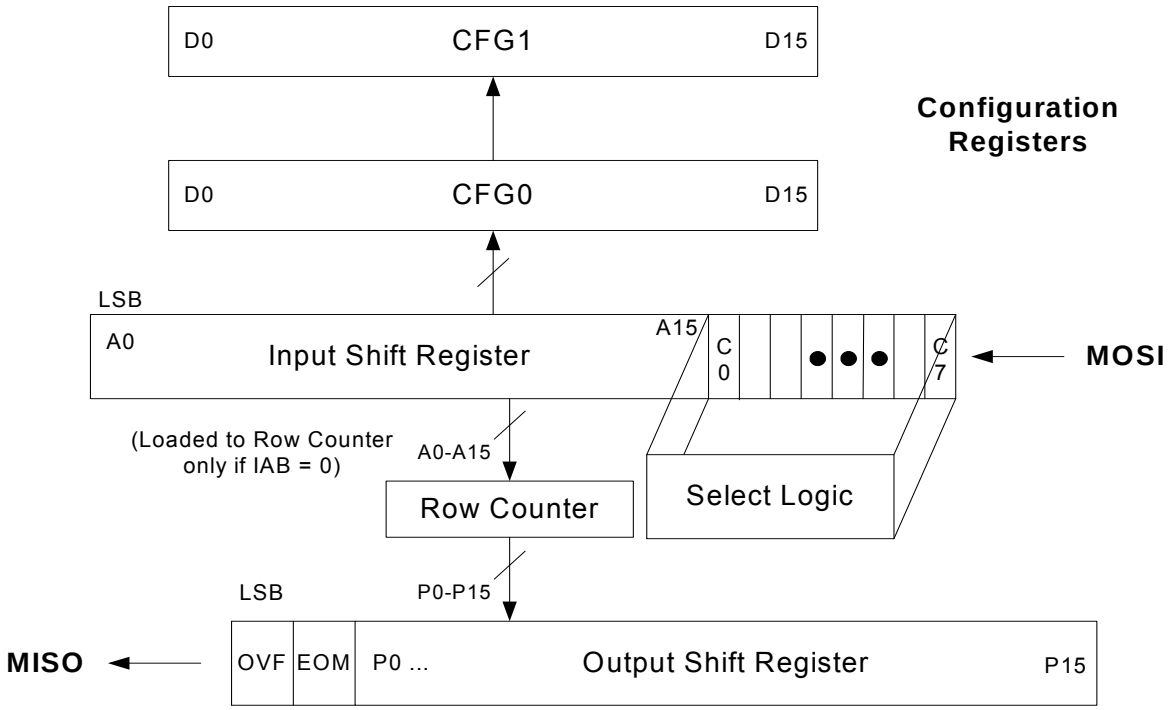
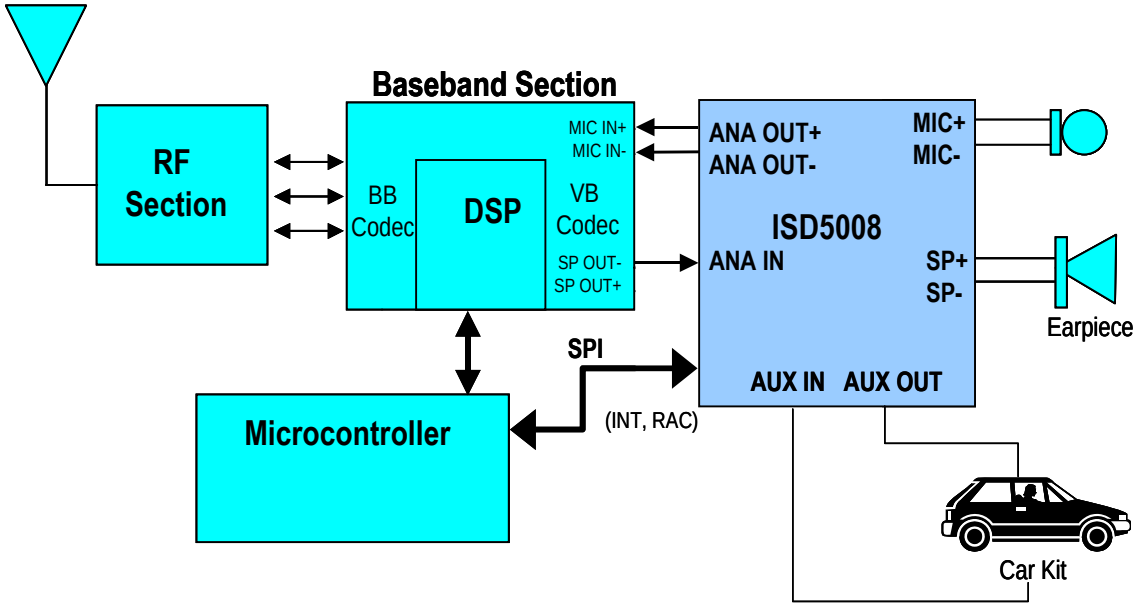

D0
CFG1
D15
Configuration Registers
D0
CFG0
D15
LSB
A0
Input Shift Register
A15
C 0
C 7
MOSI
(Loaded to Row Counter only if IAB = 0)
A0-A15
Row Counter
Select Logic
LSB
P0-P15
MISO
OVF
EOM
P0 ...
Output Shift Register
P15

FIGURE 15: TypicalDigital Cellular Phone Integration



6.5 OPERATIONAL MODES DESCRIPTION

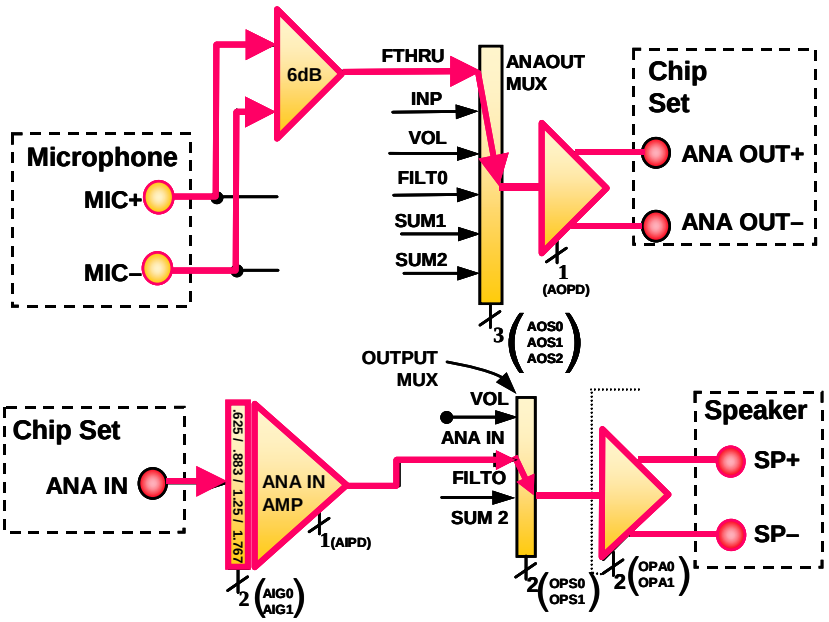
The ISD5008 can operate in many different modes. It's flexibility allows the user to configure the chip such that almost any input can mixed with any other input and then be directed to any output. The variable settings for the ANA and AUX input amplifiers plus the microphone AGC and speaker volume controls make it possible to use the device with most existing cell phone or cordless phone chip sets with no external level adjustment. Several modes will be found in most applications, however, please refer to the ISD5008 block diagram to better understand the following modes. In all cases, we are assuming that the chip has been powered up with the PU bit in the SPI control register and that a time period of T_{PUD} has elapsed after that bit was set.

6.5.1 Feed Through Mode

This mode enables the ISD5008 to connect to a base band cell phone or cordless phone chip set without affecting the audio source or destination. There are two paths involved, the transmit path and the receive path. The transmit path connects the ISD chip's microphone source through to the microphone input on the base band chip set. The receive path connects the base band chip set's speaker output through to the speaker driver on the ISD chip. This allows the ISD chip to substitute for those functions and incidentally gain access to the audio to and from the base band chip set. Figure 15 shows one possible connection to such a chip set.

Figure 16 shows the part of the ISD5008 block diagram that is used in Feed Through Mode. The rest of the chip will be powered down to conserve power. The bold lines highlight the audio paths. Note that the Microphone to ANA OUT +/– path is differential

FIGURE 16: BASIC FEED THRU MODE



To select this mode, the following control bits must be configured in the ISD5008 configuration registers. To set up the transmit path:

1. *Select the FTHRU path through the ANA OUT MUX*—Bits AOS0, AOS1 and AOS2 control the state of the ANAOUT MUX. These are the D6, D7 and D8 bits respectively of Configura tion Register 0 (CFG0) and they should all be ZERO to select the FTHRU path.
2. *Power up the ANA OUT amplifier*—Bit AOPD controls the power up state of ANA OUT. This is bit D5 of CFG0 and it should be a ZERO to power up the amplifier.

To set up the receive path:

1. *Set up the ANA IN amplifier for the correct gain*—Bits AIG0 and AIG1 control the gain settings of this amplifier. These are bits D14 and D15 respectively of CFG0. The input level at this pin determines the setting of this gain stage. Table 4 will help determine this setting. In this example we will assume that the peak signal never goes above 1 volt p-p single ended. That would enable us to use the 9dB attenuation setting, or where D14 is ONE and D15 is ZERO.
2. *Power up the ANA IN amplifier*—Bit AIPD controls the power up state of ANA IN. This is bit D13 of CFG0 and should be a ZERO to power up the amplifier.
3. *Select the ANA IN path through the OUTPUT MUX*—Bits OPS0 and OPS1 control the state of the OUTPUT MUX. These are bits D3 and D4 respectively of CFG0 and they should be set to the state where D3 is ONE and D4 is ZERO to select the ANA IN path.
4. *Power up the Speaker Amplifier*—Bits OPA0 and OPA1 control the state of the Speaker and AUX amplifiers. These are bits D1 and D2 respectively of CFG0. They should be set to the state where D1 is ONE and D2 is ZERO. This powers up the Speaker Amplifier and

configures it for it's higher gain setting for use with a piezo speaker element and also powers down the AUX output stage.

The status of the rest of the functions in the ISD5008 chip must be defined before the configuration registers settings are updated:

1. *Power down the Volume Control Element*—Bit VLPD controls the power up state of the Volume Control. This is bit D0 fo CFG0 and it should be set to a ONE to power down the statge.
2. *Power down the AUX IN amplifier*—Bit AXPD controls the power up state of the AUX IN input amplifier. This is bit D10 of CFG0 and it should be set to a ONE to power down the stage.
3. *Power down the SUM1 and SUM2 Mixer amplifiers*—Bits S1M0 and S1M1 control the SUM mixer and bits S2M0 and S2M1 control the SUM2 mixer. These are bits D7 and D8 in CFG1 and bits D5 and D6 in CFG1 respectively. All 4 bits should be set to a ONE to power down the stage.
4. *Power down the FILTER stage*—Bit FLPD controls the power up state of the FILTER stage in the device. This is bit D0 in CFG1 and should be set to a ONE to power down the stage.
5. *Power down the AGC amplifier*—Bit AGPD controls the power up state of the AGC amplifier. This is bit D0 in CFG1 and should be set to a ONE to power down the stage.
6. *Don't Care bits*—The following stages are not used in Feed Through Mode. Their bits may be set to either level. In this example we will set all the following bits to ZERO.
    a. Bit INS0, bit D9 of CFG0 controls the Input Source Mux.
    b. Bits AXG0 and AXG1 are bits D11 and D12 respectively in CFG0. They control the AUX IN amplifier gain setting.
    c. Bits FLD0 and FLD1 are bits D2 and D3 respectively in CFG1. They control the sample rate and filter band pass setting.
    d. Bit FLS0 is bit D4 in CFG1. It controls the FILTER MUX.
    e. Bits S1S0 and S1S1 are bits D9 and D10 of CFG1. They control the SUM1 MUX.
    f. Bits VOL0, VOL1 and VOL2 are bits D11, D12 and D13 of CFG1. They control the setting of the Volume Control.
    g. Bits VLS0 and VLS1 are bits D14 and D15 of CFG1. They control the Volume Control Mux.

The end result of the above set up is:

CFG0 = 0100 0100 0000 1011 (hex 4408)

and

CFG1 = 0000 0001 1110 0011 (hex 01E3)

Since both registers are being loaded, CFG0 is loaded followed by the loading of CFG1. These two registers must be loaded in this order. The internal set up for both registers will take effect synchronously with the rising edge of $\overline{SS}$.

6.5.2 Call Record

The call record mode adds the ability to record the incoming phone call. In most applications, the ISD5008 would first be set up for Feed Through Mode as described above. When the user wishes to record the incoming call, the set up of the chip is modified to add that ability. For the purpose of this explanation, we will use the 6.4 kHz sample rate during recording.

The block diagram of the ISD5008 shows that the Multilevel Storage array is always driven from the SUM2 SUMMING amplifier. The path traces back from there through the LOW PASS Filter, THE FILTER MUX, THE SUM1 SUMMING amplifier, the SUM1 MUX, then from the ANA in amplifier. Feed Through Mode has already powered up the ANA IN amp so we only need to power up and enable the path to the Multilevel Storage array from that point:

1. *Select the ANA IN path through the SUM1 MUX*—Bits S1S0 and S1S1 control the state of the SUM1 MUX. These are bits D9 and D10 respectively of CFG1 and they should be set to the state where both D9 and D10 are ZERO to select the ANA IN path.
2. *Select the SUM1 MUX input (only) to the S1 SUMMING amplifier*—Bits S1M0 and S1M1 control the state of the SUM1 SUMMING amplifier. These are bits D7 and D8 respectively of CFG1 and they should be set to the state where D7 is ONE and D8 is ZERO to select the SUM1 MUX (only) path.
3. *Select the SUM1 SUMMING amplifier path through the FILTER MUX*—Bit FLS0 controls the state of the FILTER MUX. This is bit D4 of CFG1 and it should be set to ZERO to select the SUM1 SUMMING amplifier path.
4. *Power up the LOWPASS FILTER*—Bit FLPD controls the power up state of the LOWPASS FILTER stage. This is bit D1 of CFG1 and it should be set to ZERO to power up the LOW PASS FILTER STAGE.
5. *Select the 6.4 kHz sample rate*—Bits FLD0 and FLD1 select the Low Pass filter setting and sample rate to be used during record and playback. These are bits D2 and D3 of CFG1. To enable the 6.4 kHz sample rate, D2 should be set to ONE and D3 to ZERO.
6. *Select the LOW PASS FILTER input (only) to the SUM2 SUMMING amplifier*—Bits S2M0 and S2M1 control the state of the SUM2 SUMMING amplifier. These are bits D5 and D6 respectively of CFG1 and they should be set to the state where D5 is ZERO and D6 is ONE to select the LOW PASS FILTER (only) path.

In this mode, the elements of the original PASS THROUGH mode do not change. The sections of the chip not required to add the record path remain powered down. In fact, CFG0 does not change and remains

CFG0=0100 0100 0000 1011 (hex 440B).

CFG1 changes to

CFG1=0000 0000 1100 0101 (hex 00C5).

Since CFG0 is not changed, it is only necessary to load CFG1. Note that if only CFG0 was changed, it would be necessary to load both registers.

6.5.3 Memo Record

The Memo Record mode sets the chip up to record from the local microphone into the chip's Multilevel Storage Array. A connected cellular telephone or cordless phone chip set may remain powered down and is not active in this mode. The path to be used is microphone input to AGC amplifier, then through the INPUT SOURCE MUX to the SUM1 SUMMING amplifier. From there the path goes through the FILTER MUX, the LOW PASS FILTER, the SUM2 SUMMING amplifier, then to the MULTILEVEL STORAGE ARRAY. In this instance, we will select the 5.3 kHz sample rate. The rest of the chip may be powered down.

1. *Power up the AGC amplifier*—Bit AGPD controls the power up state of the AGC amplifier. This is bit D0 of CFG1 and should be set to ZERO to power up the stage.
2. *Select the AGC amplifier through the INPUT SOURCE MUX*—Bit INS0 controls the state of the INPUT SOURCE MUX. This is bit D9 of CFG0 and should be set to a ZERO to select the AGC amplifier.
3. *Select the INPUT SOURCE MUX (only) to the SUM1 SUMMING amplifier*—Bits S1M0 and S1M1 control the state of the SUM1 SUMMING amplifier. These are bits D7 and D8 respectively of CFG1 and they should be set to the state where D7 is ZERO and D8 is ONE to select the INPUT SOURCE MUX (only) path.
4. *Select the SUM1 SUMMING amplifier path through the FILTER MUX*—Bit FLS0 controls the state of the FILTER MUX. This is bit D4 of CFG1 and it should be set to ZERO to select the SUM1 SUMMING amplifier path.
5. *Power up the LOW PASS FILTER*—Bit FLPD controls the power up state of the LOW PASS FILTER stage. This is bit D1 of CFG1 and it should be set to ZERO to power up the LOW PASS FILTER stage.
6. *Select the 5.3 kHz sample rate*—Bits FLD0 and FLD1 select the Low Pass filter setting and sample rate to be used during record and playback. These are bits D2 and D3 of CFG1. To enable the 5.3 kHz sample rate, D2 should be set to ZERO and D3 set to ONE.
7. *Select the LOW PASS FILTER input (only) to the SUM2 SUMMING amplifier*—Bits S2M0 and S2M1 control the state of the SUM2 SUMMING amplifier. These bits are D5 and D6 respectively of CFG1 and they should be set to the state where D5 is ZERO and D6 is ONE to select the LOW PASS FILTER (only) path.

To set up the chip for Memo Record, the configuration registers are set up as follows:

CFG0=0010 0100 0010 0001 (hex 2421).

CFG1=0000 0001 0100 1000 (hex 0148).

Only those portions necessary for this mode are powered up.

6.5.4 Memo and Call Playback

This mode sets the chip up for local playback of messages recorded earlier. The playback path is from the MULTILEVEL STORAGE ARRAY to the FILTER MUX, then to the LOW PASS FILTER stage. From there the audio path goes through the SUM2 SUMMING amplifier to the VOLUME MUX, through the VOLUME CONTROL then to the SPEAKER output stage. We will assume that we are driving a pizeo speaker element. This audio was previously recorded at 8 kHz. All unnecessary stages will be powered down.

1. *Select the MULTILEVEL STORAGE ARRAY path through the FILTER MUX*—Bit FLS0 controls the state of the FILTER MUX. This is bit D4 of CFG1 and should be set to ONE to select the MULTILEVEL STORAGE ARRAY.
2. *Power up the LOW PASS FILTER*—Bit FLPD controls the power up of the LOW PASS FILTER stage. This is bit D1 of CFG1 and it should be set to ZERO to power up the LOW PASS FILTER stage.
3. *Select the 8.0 kHz sample rate*—Bits FLD0 and FLD1 select the Low Pass filter setting and sample rate to be used during record and playback. These are bits D2 and D3 of CFG1. To enable the 8.0 kHz sample rate, D2 and D3 must be set to ZERO.
4. *Select the LOW PASS FILTER input (only) to the S2 SUMMING amplifier* —Bits S2M0 and S2M1 control the state of the SUM2 SUMMING amplifier. These are bits D5 and D6 respectively of CFG1 and they should be set to the state where D5 is ZERO and D6 is ONE to select the LOW PASS FILTER (only) path.
5. *Select the SUM2 SUMMING amplifier path through the VOLUME MUX*—Bits VLS0 and VLS1 control the state VOLUME MUX. These bits are D14 and D15, respectively of CFG1. They should be set to the state where D14 is ONE and D15 is ZERO to select the SUM2 SUMMING amplifier.
6. *Power up the VOLUME CONTROL LEVEL*—Bit VLPD controls the power up state of the VOLUME CONTROL attenuator. This is bit D0 of CFG0. This bit should be set a ZERO to power up the VOLUME CONTROL.
7. *Select a VOLUME CONTROL LEVEL*—Bits VOL0, VOL1 and VOL2 control the state fo the VOLUME CONTROL LEVEL. Theses are bits D11, D12 and D13, repectively of CFG1. A binary count of 000 through 111 controls the amount of attenuation throught that state. In most cases, the software will select an attenuation level according to the desires of the current users of the product. In this example, we will assume the user wants an attenuation of –12 dB. For that setting, D11 should be set to ONE, D12 should be set to ONE, and D13 shoulde be set to ZERO.
8. *Select the VOLUME CONTROL path through the OUTPUT MUX*—Bits OPS0 and OPS1 control the state of the OUTPUT MUX. These are bits D3 and D4, respectively of CFG0. They should be set to the state where D3 and D4 are ZERO to select the VOLUME CONTROL.
9. *Power up the SPEAKER amplifier and select the HIGH GAIN mode*—Bits OPA0 and OPA1 control the state of the speaker (SP+ and SP-) and AUX OUT outputs. These are bits D1 and D2 of CFG0. They should be set to the state where D1 is ONE and D2 is ZERO to power up the speaker outputs in the HIGH GAIN mode and to power down the AUX OUT.

To set up the chip for Memo or Call Playback, the configuration registers are set up as follows:

CFG0=0010 0100 0010 0010 (hex 2422).

CFG1=0101 1001 1101 0001 (hex 59D1).

Only those portions necessary for this mode are powered up.

7 TIMING DIAGRAMS

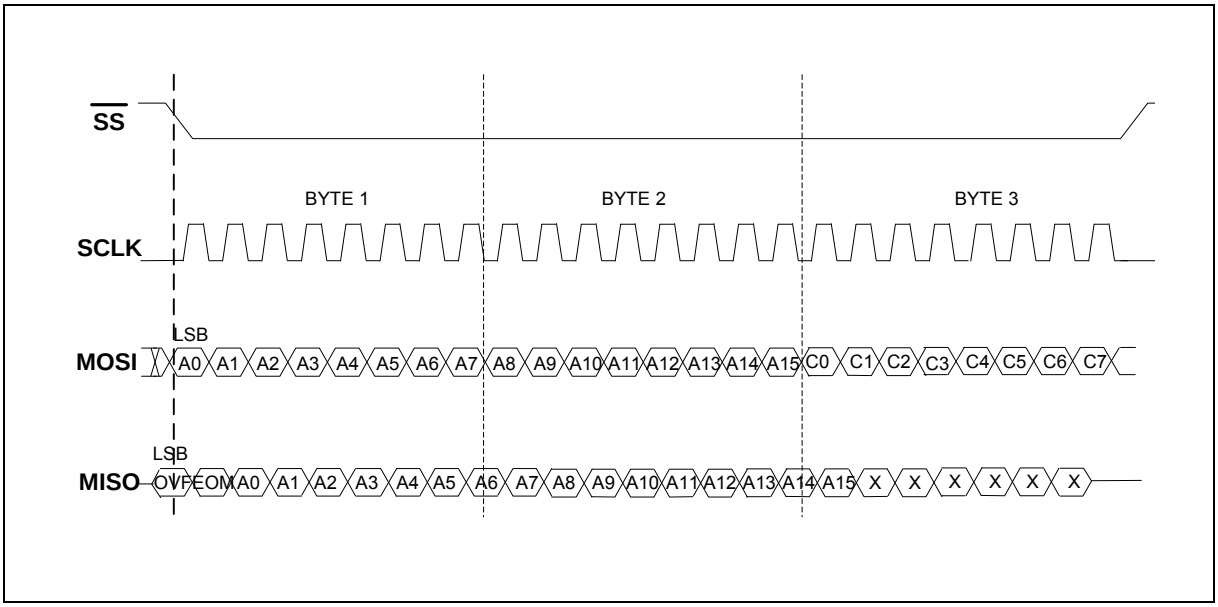


FIGURE 17: 24-BIT SPI COMMAND FORMAT

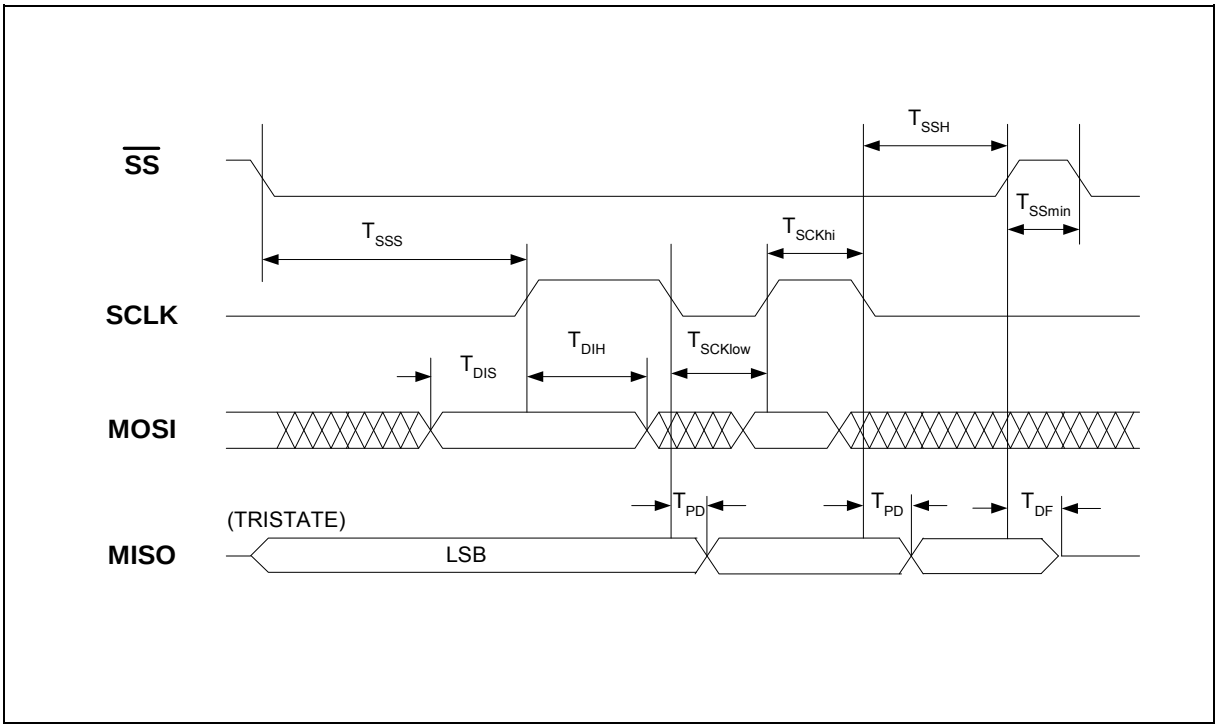


FIGURE 18: SPI TIMING DIAGRAM

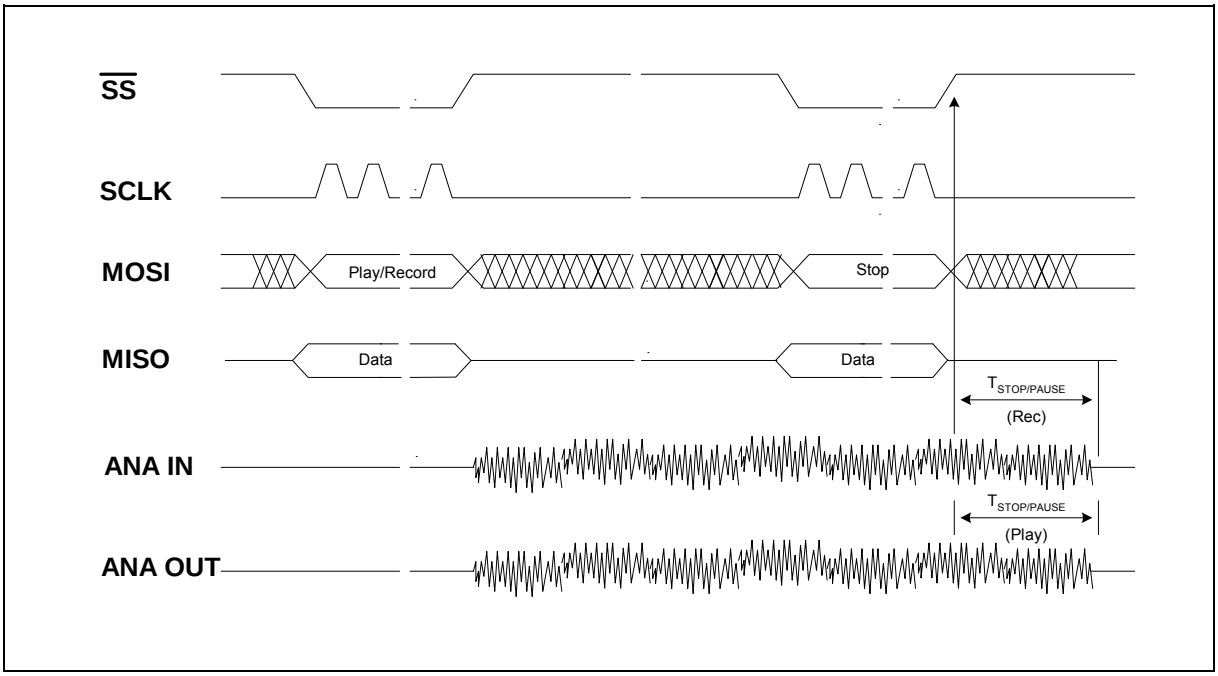


FIGURE 19: PLAYBACK/RECORD AND STOP CYCLE

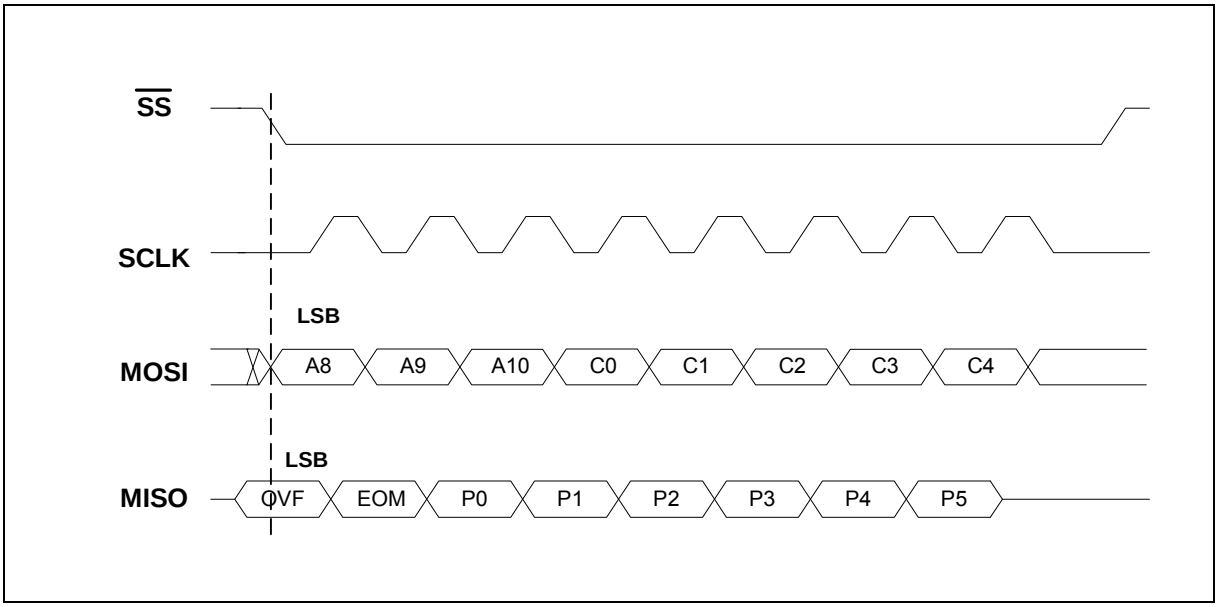


FIGURE 20: 8-Bit SPI Command Format

8 ABSOLUTE MAXIMUM RATINGS [1]

TABLE 10: ABSOLUTE MAXIMUM RATINGS (PACKAGED PARTS)

CONDITIONS	VALUES
Junction temperature	150ºC
Storage temperature range	-65ºC to +150ºC
Voltage applied to any pin	(V_{SS} – 0.3V) to (V_{CC} + 0.3V)
Voltage applied to MOSI, SCLK, $\overline{INT}$, RAC and $\overline{SS}$ pins (Input current limited to ±20mA)	(V_{SS} – 1.0V) to 5.5V
Lead temperature (Soldering – 10sec)	300ºC
V_{CC} – V_{SS}	-0.3V to +7.0V

TABLE 11: ABSOLUTE MAXIMUM RATINGS (DIE)

CONDITIONS	VALUES
Junction temperature	150ºC
Storage temperature range	-65ºC to +150ºC
Voltage applied to MOSI, SCLK, $\overline{INT}$, RAC and $\overline{SS}$ pins (Input current limited to ±20mA)	(V_{SS} – 0.3V) to 5.5V
V_{CC} – V_{SS}	-0.3V to +7.0V

Note: [1] Stresses above those listed may cause permanent damage to the device. Exposure to the absolute maximum ratings may affect device reliability and performance. Functional operation is not implied at these conditions.

8.1 OPERATING CONDITIONS

TABLE 12: OPERATING CONDITIONS (PACKAGED PARTS)

CONDITION	VALUE
Commercial operating temperature range (Case temperature)	0ºC to +70ºC
Extended operating temperature (Case temperature)	-20ºC to +70ºC
Industrial operating temperature (Case temperature)	-40ºC to +85ºC
Supply voltage (V_{CC}) [1]	+2.7V to 3.3V
Ground voltage (V_{SS}) [2]	

TABLE 13: OPERATING CONDITIONS (DIE)

CONDITION	VALUE
Commercial operating temperature range	0ºC to +50ºC
Supply voltage (V_{CC}) [1]	+2.7V to +3.3V
Ground voltage (V_{SS}) [2]	0V

[1] $V_{CC} = V_{CCA} = V_{CCD}$

[2] $V_{SS} = V_{SSA} = V_{SSD}$

9 ELECTRICAL CHARACTERISTICS

9.1 GENERAL PARAMETERS

TABLE 14: GENERAL PARAMETERS

PARAMETERS	SYMBOLS	MIN(2)	TYP(1)	MAX(2)	UNITS	CONDITIONS
Input Low Voltage	V_{IL}			V_{CC} x 0.2	V	
Input High Voltage	V_{IH}	V_{CC} x 0.8			V	
Output Low Voltage	V_{OL}			0.4	V	I_{OL} = 10 µA
RAC, $\overline{INT}$ Output Low Voltage	V_{OL1}			0.4	V	I_{OL} = 1 mA
Output High Voltage	V_{OH}	V_{CC} - 0.4			V	I_{OH} = -10 µA
Operating Current :	I_{CC}					
- Playback			15		mA	No load (3)
- Record			25		mA	No load (3)
- Feedthru			12		mA	No load (3)
Standby Current	I_{SB}		1	10	µA	(3) (4)
Input Leakage Current	I_{IL}			±1	µA	
MISO Tristate Current	I_{HZ}	30	1	10	µA	

1. Typical values @ T_A = 25° and 3.0V.
2. All Min/Max limits are guaranteed by Nuvoton via electronical testing or characterization. Not all specifications are 100 percent tested.
3. V_{CCA} and V_{CCD} summed together.
4. $\overline{SS}$ = V_{CCA} = V_{CCD}, XCLK = MOSI = V_{SSA} = V_{SSD} and all other pins floating.

9.2 TIMING PARAMETERS

TABLE 15: TIMING PARAMETERS

CHARACTERISTICS	SYMBOLS	MIN(2)	TYP(1)	MAX(2)	UNITS	CONDITIONS
Sampling Frequency	F_S		8.0		kHz	(4)
			6.4		kHz	(4)
			5.3		kHz	(4)
			4.0		kHz	(4)
Filter Pass Band	F_{CF}					3-dB Roll-Off Point(3)(7)
8.0 kHz (sample rate)			3.4		kHz	
6.4 kHz (sample rate)			2.7		kHz	
5.3 kHz (sample rate)			2.3		kHz	
4.0 kHz (sample rate)			1.7		kHz	
Record Duration	T_{REC}					
8.0 kHz (sample rate)			4		min	(6)
6.4 kHz (sample rate)			5		min	(6)
5.3 kHz (sample rate)			6		min	(6)
4.0 kHz (sample rate)			8		min	(6)
Playback Duration	T_{PLAY}					
8.0 kHz (sample rate)			4		min	(6)
6.4 kHz (sample rate)			5		min	(6)
5.3 kHz (sample rate)			6		min	(6)
4.0 kHz (sample rate)			8		min	(6)
Power-Up Delay	T_{PUD}					
8.0 kHz (sample rate)			25		msec	
6.4 kHz (sample rate)			31.25		msec	
5.3 kHz (sample rate)			37.5		msec	
4.0 kHz (sample rate)			50		msec	
Stop or Pause	$T_{STOP\ or\ PAUSE}$					
8.0 kHz (sample rate)			50		msec	
6.4 kHz (sample rate)			62.5		msec	
5.3 kHz (sample rate)			75		msec	
4.0 kHz (sample rate)			100		msec	

RAC Clock Period 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)	T_{RAC}		 200 250 300 400		 msec msec msec msec	 (9) (9) (9) (9)
RAC Clock Low Time 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)	T_{RACLO}		 25 31.25 37.5 50		 msec msec msec msec	
RAC Clock Periond in Messge Cueing Mode 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)	T_{RACM}		 125 156.3 187.5 250		 µsec µsec µsec µsec	
RAC Clock Low Time in Message Cueing Mode 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)	T_{RACML}		 15.63 19.53 23.44 31.25		 µsec µsec µsec µsec	
Total Harmonic Distortion ANA IN to ARRAY, ARRAY to SPKR	THD		1	2	%	@ 1 kHz at 0TLP, sample rate = 5.3 kHz

9.3 ANALOG PARAMETERS

TABLE 16: ANALOG PARAMETERS

CHARACTERISTICS	SYMBOL	MIN (2)	TYP (1)	MAX (2)	UNITS	CONDITIONS
MICROPHONE INPUT (14)						
MIC+/- Input Voltage	$V_{MIC+/-}$	3		300	mV	Peak-to-Peak (4)(8)
MIX+/- input reference transmission level point (0TLP)	$V_{MIC\ (0TLP)}$		208		mV	Peak-to-Peak (4)(10)
Gain from MIC+/- input to ANA OUT	A_{MIC}	5.5	6.0	6.5	dB	1kHz at $V_{MIC\ (0TLP)}$ (4)

MIC+/- Gain Tracking	$A_{MIC\ (GT)}$		±0.1		dB	1kHz, +3 to 40 dB 0TLP Input
Microphone input resistance	R_{MIC}	5	10	15	Ω	MIC- and MIC+ pins
Microphone AGC Amplifier Range	A_{AGC}	6		40	dB	Over 3-300 mV Input Range
ANA IN [9]						
ANA IN Input Voltage	$V_{ANA\ IN}$			1.6	V	Peak-to-Peak (6dB gain setting)
ANA IN (0TLP) Input Voltage	$V_{ANA\ IN\ (0TLP)}$		1.11		V	Peak-to-Peak (6dB gain setting) [10]
Gain from ANA IN to SP+/-	$A_{ANA\ IN\ (SP)}$		6 to 15		dB	4 Steps of 3 dB
Gain from ANA IN to AUX OUT	$A_{ANA\ IN\ (AUX\ OUT)}$		-4 to +5		dB	4 Steps of 3 dB
ANA IN Gain Accuracy	$A_{ANA\ IN\ (GA)}$	-0.5		+0.5	dB	[11]
ANA IN Gain Tracking	$A_{ANA\ IN\ (GT)}$		±0.1		dB	1000 Hz, +3 to -40dB 0TLP Input, 6dB setting
ANA IN Input Resistance	$R_{ANA\ IN}$		60 to 102		kΩ	See Ra in Figure 2
AUX IN [9]						
AUX IN Iput Voltage	$V_{AUX\ IN}$			1.0	V	Peak-to-Peak (0dB gain setting)
AUX IN (0TLP) Input Voltage	$V_{AUX\ IN\ (0TLP)}$		694.2		mV	Peak-to-Peak (0dB gain setting) [10]
Gain from AUX IN to ANA OUT	$A_{AUX\ IN\ (ANA\ OUT)}$		0 to 9		dB	4 Steps of 3dB
AUX IN Gain Accuracy	$A_{AUX\ IN\ (GA)}$	-0.5		+0.5	dB	[11]
AUX IN Gain Tracking	$A_{AUX\ IN\ (GT)}$		±0.1		dB	1000 Hz, +3 to -40dB 0TLP Input, 0dB setting
AUX IN Input Resistance	$R_{AUX\ IN}$		21 to 40		kΩ	See Ra in Figure 3
SPEAKER OUTPUTS [9]						
SP+/- Output Voltage (High Gain setting)	V_{SPHG}			3.6	V	Peak-to-Peak, differential load = 150 Ω; OPA1, OPA0 = 01
SP+/- Output Load Imp. (Low Gain)	R_{SPLG}	8			Ω	OPA1, OPA0 = 10
SP+/- Output Load Imp. (High Gain)	R_{SPHG}	70			Ω	OPA1, OPA0 = 01
SP+/- Output Load Cap.	C_{SP}			100	pF	
SP+/- Output Bias Voltage (analog ground)	V_{SPAG}		1.2		VDC	

Speaker Output DC Offset	V_{SPDCO}	-100		100	mVDC	With ANA IN to Speaker, ANA IN AC coupled to V_{SSA}
ANA IN to SP+/- Idle Channel Noise	$ICN_{ANA\ IN/(SP+/-)}$			-65	dB	Speaker load = 150 Ω (12)(13)
SP+/- to ANA OUT Cross Talk	$C_RT_{(SP+/-)\ ANA\ OUT}$			-65	dB	1kHz 0TLP input to ANA IN, with MIC+/ and AUX IN AC coupled to V_{SSA}, and measured at ANA OUT feedthrough mode (12)
Power Supply Rejection Ratio	PSRR		-50		dB	Measured with a 1kHz,100 mVpp sine wave input at V_{CCA} and V_{CCD} pins
Frequency Response (300-3400 Hz)	F_R	-0.25		+0.25	dB	With 0TLP input to ANA IN, 6dB setting (12)
Power Output (Low Gain Setting)	P_{OUTLG}	23.5			mW RMS	Differential load at 8 Ω
SINAD ANA IN to SP+/-	SINAD	62.5			dB	0TLP ANA IN input minimum gain, 150 Ω load (12)(13)
ANA OUT (9)						
SINAD MIC IN to ANA Out+/-	SINAD	62.5			dB	Load = 5 kΩ (12)(13)
SINAD AUX IN to ANA OUT+/-	SINAD	62.5			dB	Load = 5 kΩ (12)(13)
Idle Channel Noise – Microphone	$ICN_{MIC/ANA\ OUT}$			-65	dB	Load = 5 kΩ (12)(13)
Idle Channel Noise – AUX IN (0 to 9 dB)	$ICN_{AUX\ IN\ /NA\ OUT}$			-65	dB	Load = 5 kΩ (12)(13)
Power Supply Rejection Ratio	$PSRR_{(ANA\ OUT)}$		-50		dB	Measured with a 1kHz, 100mVpp sine wave to V_{CCA}, V_{CCD} pins
ANA OUT+ and ANA OUT-	V_{BIAS}		1.2		VDC	Inputs AC coupled to V_{SSA}
ANA OUT+ and ANA OUT-	V_{OFFSET}	-100		+100	mVDC	Inputs AC coupled to V_{SSA}
Minimum Load Impedence	R_L	5			kΩ	Differential Load
Frequency Response (300-3400 Hz)	F_R	-0.25		+0.25	dB	0TLP input to MIC+/- in feedthrough mode. 0TLP input to AUX IN in feedthrough mode (12)

ANA OUT to SP+/- Cross Talk	$C_RT_{ANA\ OUT/(SP+/-)}$			-65	dB	1kHz 0TLP output from ANA OUT, with ANA IN AC coupled to V_{SSA}, and measured at SP+/- (12)
ANA OUT to AUX OUT Cross Talk	$C_RT_{ANA\ OUT/AUX\ OUT}$			-65	dB	1kHz 0TLP output from ANA OUT, with ANA IN AC coupled to V_{SSA}, and measured at AUX OUT (12)
AUX OUT (9)						
AUX OUT – Maximum Output Swing	$V_{AUX\ OUT}$			1.0	Vpp	5 kΩ Load
Minimum Load Impedence	R_L	5			kΩ	
Maximum Load Capacitance	C_L			100	pF	
AUX OUT	V_{BIAS}		1.2		VDC	
SINAD – ANA IN to AUX OUT	SINAD	62.5			dB	0TLP ANA IN input, minimum gain, 5k load (12)(13)
Idle Channel Noise – ANA IN to AUX OUT	$ICN_{(AUX\ OUT)}$			-65	dB	Load = 5 kΩ (12)(13)
AUX OUT to ANA OUT Cross Talk	$C_RT_{AUX\ OUT/ANA\ OUT}$			-65	dB	1 kHz 0TLP input to ANA IN, with MIC +/- and AUX IN AC coupled to V_{SSA}, and measured at SP+/-, load = 5 kΩ. Referenced to nominal 0TLP @ output
VOLUME CONTROL (9)						
Output Gain	A_{OUT}		-28 to 0		dB	8 Steps of 4 dB, referenced to output
Gain Accuracy		-0.5		0.5	dB	ANA IN = 1 kHz 0TLP, 6dB Gain setting

Notes:

1. Typical values: TA = 25°C and Vcc = 3.0V.
2. All min/max limits are guaranteed by ISD via electrical testing or characterization. Not all specifications are 100 percent tested.
3. Low-frequency cut off depends upon the value of external capacitors (see Pin Descriptions).
4. Differential input mode. Nominal differential input is 208 mVp-p. (0 dBm0)
5. Sampling frequency can vary as much as –6/+4 percent over the industrial temperature and voltage ranges. For greater stability, an external clock can be utilized (see Pin Descriptions). Sampling frequency will be accurate within ±1% for 5.3kHz, and ±5% for 4.0, 6.4 and 8.0 kHz sampling rates at room temperature.

6. Playback and Record Duration can vary as much as –6/+4 percent over the industrial temperature and voltage ranges. For greater stability, an external clock can be utilized (See Pin Descriptions). Playback and record durations are accurate within ±1% for 5.3kHz, and ±5% for 4.0, 6.4 and 8.0kHz sampling rates at room temperature.
7. Filter specification applies to the low pass filter. Therefore, from input to output, expect a 6 dB drop by nature of passing through the filter twice.
8. For optimal signal quality, this maximum limit is recommended.
9. When a record command is sent, $T_{RAC} = T_{RAC} + T_{RACLO}$ on the first row addressed.
10. The maximum signal level at any input is defined as 3.17dB higher than the reference transmission level point. (0TLP) This is the point where signal clipping may begin.
11. Measured at 0TLP point for each gain setting. See Table 4 and Table 5.
12. 0TLP is the reference test level through inputs and outputs. See Table 4 and Table 5.
13. Referenced to 0TLP input at 1kHz, measured over 300 to 3,400 Hz bandwidth.
14. For die, only typical values from Analog Parameters are applicable.

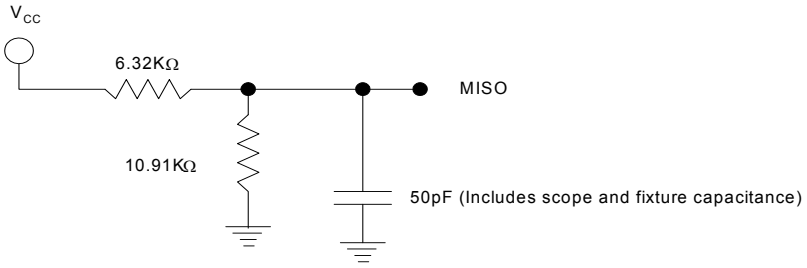
9.4 SPI AC PARAMETERS (1)

TABLE 17: SPI AC PARAMETERS

CHARACTERISTIC	SYMBOL	MIN	MAX	UNITS	CONDITIONS
$\overline{SS}$ Setup Time	T_{SSS}	500		nsec	
$\overline{SS}$ Hold Time	T_{SSH}	500		nsec	
Data in Setup Time	T_{DIS}	200		nsec	
Data in Hold Time	T_{DIH}	200		nsec	
Output Delay	T_{PD}		500	nsec	
Output Delay to hiZ	T_{DF}		500	nsec	(2)
$\overline{SS}$ HIGH	T_{SSmin}	1		µsec	
SCLK High Time	T_{SCKhi}	400		nsec	
SCLK Low Time	T_{SCKlow}	400		nsec	
CLK Frequency	F_0		1,000	kHz	

Notes:

1. Typical values @ T_A = 25° and V_{CC} = 3.0V. Timing measured at 50 percent of the V_{CC} level.
2. Tristate test condition.



10 TYPICAL APPLICATION CIRCUIT

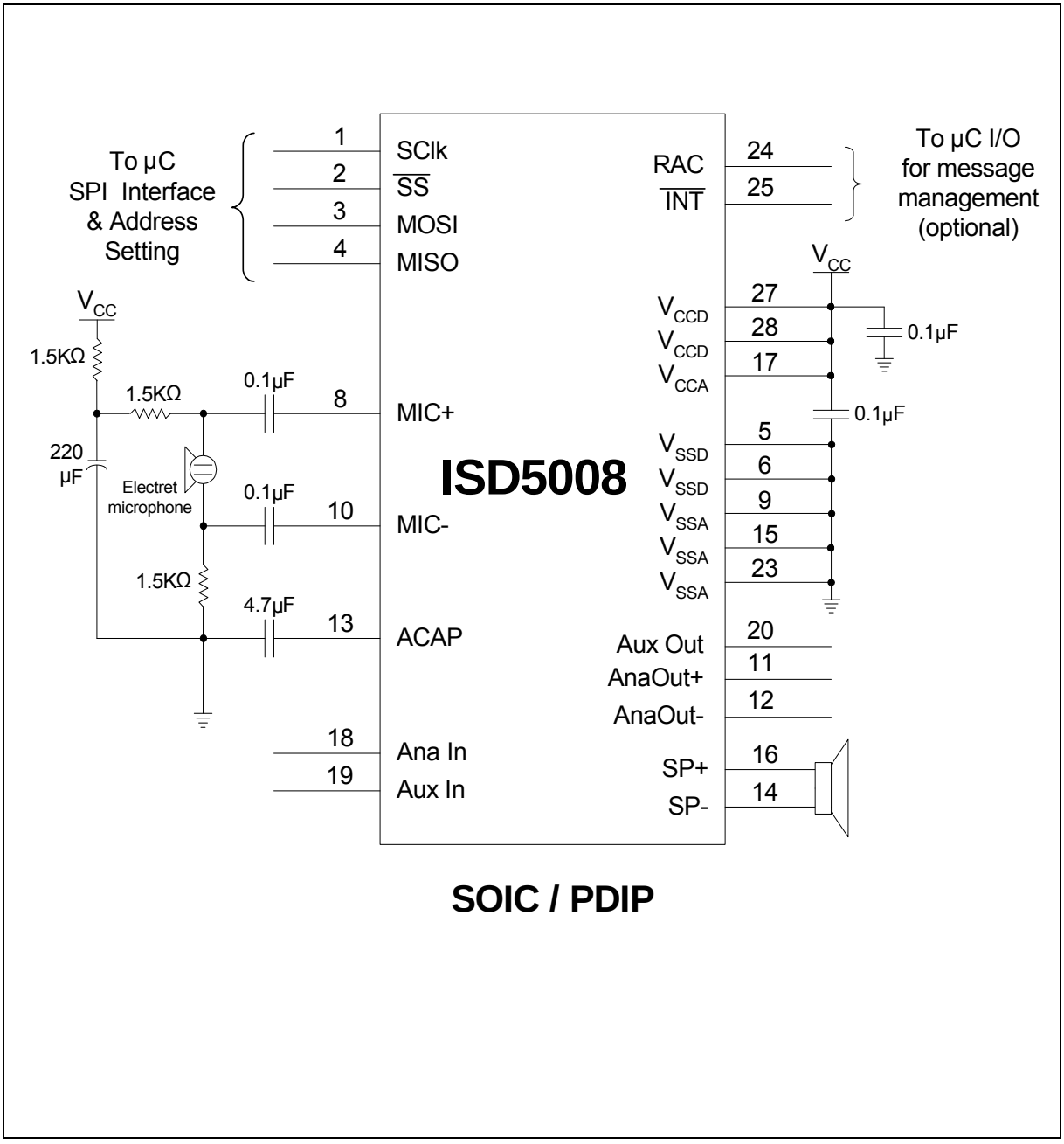
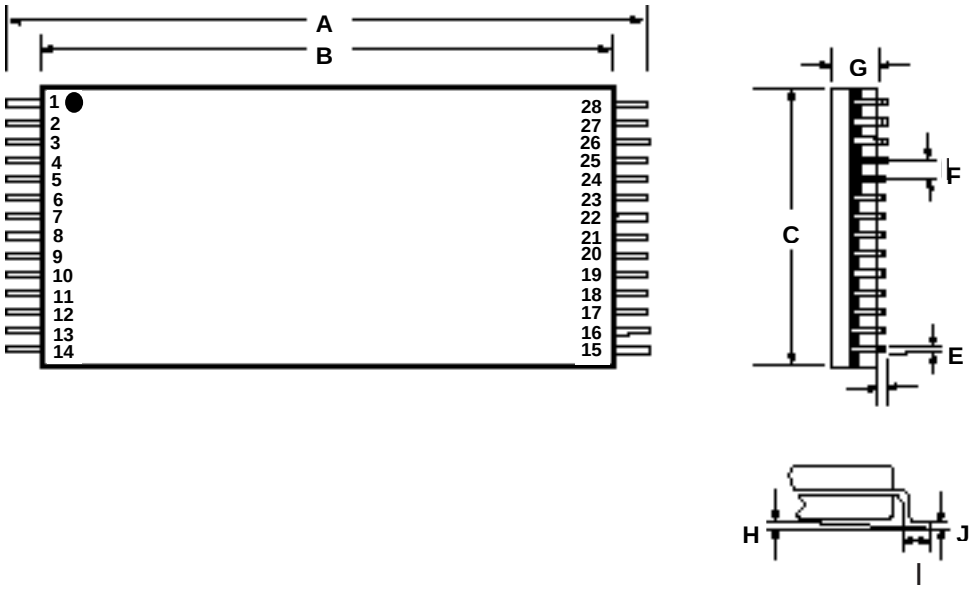


FIGURE 21: Recording via Microphone

11 PACKAGE DRAWING AND DIMENSIONS

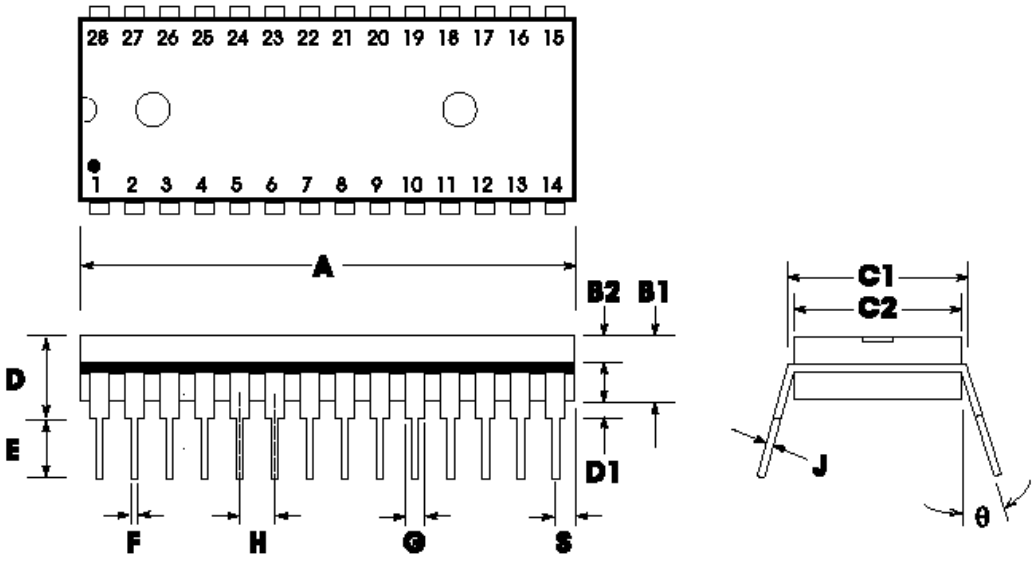
11.1 28-LEAD LEAD 8X13.4MM PLASTIC THIN SMALL OUTLINE PACKAGE (TSOP) TYPE 1



	INCHES			MILLIMETERS		
	Min	Nom	Max	Min	Nom	Max
A	0.701	0.706	0.711	17.81	17.93	18.06
B	0.097	0.101	0.104	2.46	2.56	2.64
C	0.292	0.296	0.299	7.42	7.52	7.59
D	0.005	0.009	0.0115	0.127	0.22	0.29
E	0.014	0.016	0.016	0.35	0.41	0.48
F		0.050			1.27	0
G	0.400	0.406	0.410	10.16	10.31	10.41
H	0.024	0.032	0.040	0.61	0.81	1.02

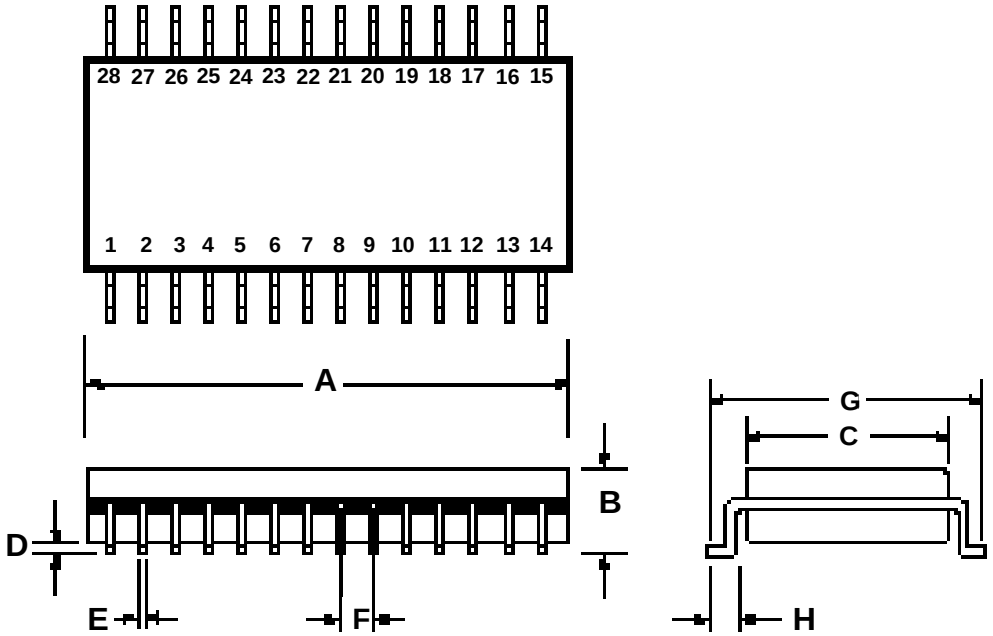
Note: Lead coplanarity to be within 0.004 inches.

11.2 28-LEAD 600 MIL PLASTIC DUAL INLINE PACKAGE (PDIP)



	INCHES			MILLIMETERS		
	Min	Nom	Max	Min	Nom	Max
A	1.445	1.450	1.455	36.70	36.83	36.96
B1		0.150			3.81	
B2	0.065	0.070	0.075	1.65	1.78	1.91
C1	0.600		0.625	15.24		15.88
C2	0.530	0.540	0.550	13.46	13.72	13.97
D			0.19			4.83
D1	0.015			0.38		
E	0.125		0.135	3.18		3.43
F	0.015	0.018	0.022	0.38	0.46	0.56
G	0.055	0.060	0.065	1.40	1.52	1.62
H		0.100			2.54	
J	0.008	0.010	0.012	0.20	0.25	0.30
S	0.070	0.075	0.080	1.78	1.91	2.03
θ	0°		15°	0°		15°

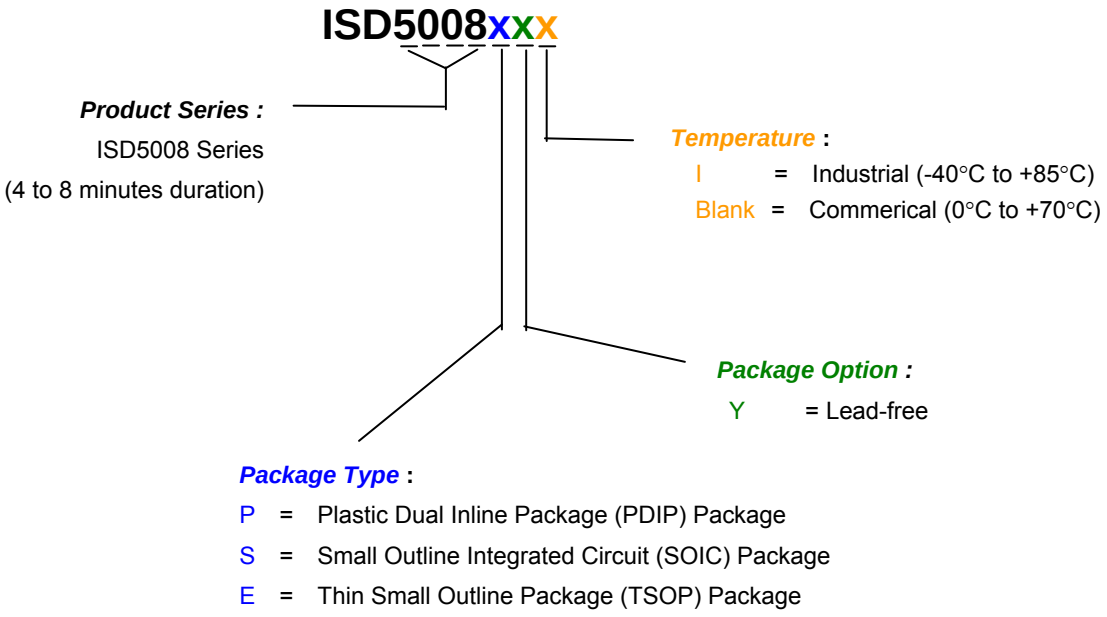
11.3 28-LEAD 300 MIL PLASTIC SMALL OUTLINE IC (SOIC)



	INCHES			MILLIMETERS		
	Min	Nom	Max	Min	Nom	Max
A	0.701	0.706	0.711	17.81	17.93	18.06
B	0.097	0.101	0.104	2.46	2.56	2.64
C	0.292	0.296	0.299	7.42	7.52	7.59
D	0.005	0.009	0.0115	0.127	0.22	0.29
E	0.014	0.016	0.019	0.35	0.41	0.48
F		0.050			1.27	
G	0.400	0.406	0.410	10.16	10.31	10.41
H	0.024	0.032	0.040	0.61	0.81	1.02

Note: Lead coplanarity to be within 0.004 inches.

12 ORDERING INFORMATION



When ordering, please refer to the above valid part number scheme that are supported in volume for this product series. Contact the local Nuvoton Sales Representative or Distributor for availability information.

For the latest product information, access Nuvoton's worldwide website at http://www.Nuvoton-usa.com

13 VERSION HISTORY

VERSION	DATE	DESCRIPTION
0	Before 2004	Initial issue
1.0	April, 2004	• Reformat the document. • Add application diagram. • Revise die info. • Revise ordering information.
1.1	Feb, 2008	• Update Opcode • Update power-up sequence. • Update pin description of RAC and INT.

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