# ISD ChipCorder® ISD5100 Series Datasheet

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## 1. GENERAL DESCRIPTION

The ISD5100 ChipCorder<sup>®</sup> Series provide high quality, fully integrated, single-chip Record/Playback solutions for 1- to 16-minute messaging applications that are ideal for use in cellular phones, automotive communications, GPS/navigation systems and other portable products. The ISD5100 Series products are an enhancement of the ISD5000 architecture, providing: 1) the I<sup>2</sup>C serial port - address, control and duration selection are accomplished through an I<sup>2</sup>C interface to minimize pin count (ONLY two control lines required); 2) the capability of storing digital data, in addition to analog data. This feature allows customers to store phone numbers, system configuration parameters and message address locations for message management capability; 3) Various internal circuit blocks can be individually powered-up or -down for power saving.

The ISD5100 Series include:

- ISD5116 from 8 to 16 minutes
- ISD5108 from 4 to 8 minutes
- ISD5104 from 2 to 4 minutes
- ISD5102 from 1 to 2 minutes

Analog functions and audio gating have also been integrated into the ISD5100 Series products to allow easy interface with integrated digital cellular chip sets on the market. Audio paths have been designed to enable full duplex conversation record, voice memo, answering machine (including outgoing message playback) and call screening features. This product enables playback of messages while the phone is in standby, AND both simplex and duplex playback of messages while on a phone call.

Additional voice storage features for digital cellular phones include: 1) a personalized outgoing message can be sent to the person by getting caller-ID information from the host chipset, 2) a private call announce while on call can be heard from the host by giving caller-ID on call waiting information from the host chipset.

Logic Interface Options of 2.0V and 3.0V are supported by the ISD5100 Series to accommodate portable communication products (2.0- and 3.0-volt required).

Like other ChipCorder<sup>®</sup> products, the ISD5100 Series integrate the sampling clock, anti-aliasing and smoothing filters, and the multi-level storage array on a single-chip. For enhanced voice features, the ISD5100 Series eliminate external circuitry by 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 included, providing a flexible interface for multiple applications.

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

## 2. FEATURES

### **Fully-Integrated Solution**

- Single-chip voice record/playback solution
- Dual storage of digital and analog data
- Durations

Device	ISD5102	ISD5104	ISD5108	ISD5116
Duration	1 to 2 minutes	2 to 4 minutes	4 to 8 minutes	8 to 16 minutes

## Low Power Consumption

- Supply Voltage
  - Commercial Temperature = +2.7V to +3.3V
  - Industrial Temperature = +2.7V to +3.3V (+2.7V to +3.6V for ISD5108 only)
- Supports 2.0V and 3.0V interface logic
- Operating Current:
  - ICC Play = 15 mA (typical)
  - $\blacktriangleright$  I<sub>CC Rec</sub> = 30 mA (typical)
  - ICC Feedthrough = 12 mA (typical)
- Standby Current:
  - >  $I_{SB} = 1 \mu A$  (typical)
- Most stages can be individually powered down to minimize power consumption

#### **Enhanced Voice Features**

- One or two-way conversation record
- One or two-way message playback
- Voice memo record and playback
- Private call screening
- In-terminal answering machine
- Personalized outgoing message
- Private call announce while on call

#### **Digital Memory Features**

Device	ISD5102	ISD5104	ISD5108	ISD5116
Storage	Up to 512Kb	Up to 1 Mb	Up to 2 Mb	Up to 4 Mb

• Storage of phone numbers, system configuration parameters and message address table in some application

### Easy-to-use and Control

- No compression algorithm development required
- User-controllable sampling rates
- Programmable analog interface
- Standard & Fast mode I<sup>2</sup>C serial interface (100kHz 400 kHz)
- Fully addressable for multiple messages

## **High Quality Solution**

- High quality voice and music reproduction
- Nuvoton's standard 100-year message retention (typical)
- 100K record cycles (typical) for analog data
- 10K record cycles (typical) for digital data

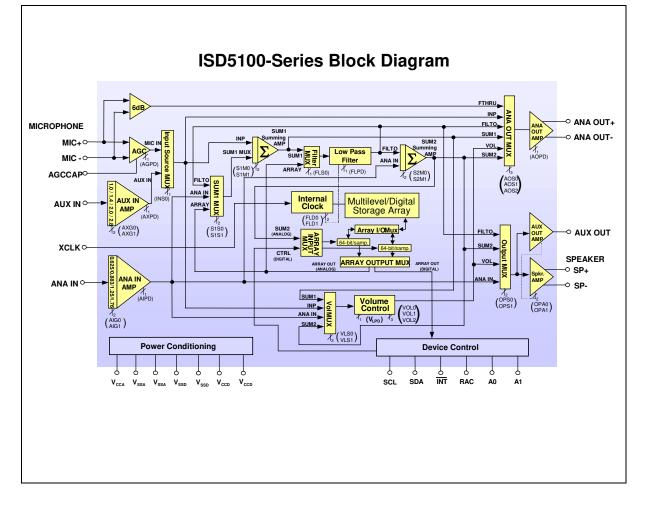
### Options

- Available in die form ,SOIC and PDIP (ISD5116 Only)
- Temperature: Commercial Packaged (0 to +70°C) & die (0 to +50°C); Industrial (-40 to +85°C)
- Pb-free package

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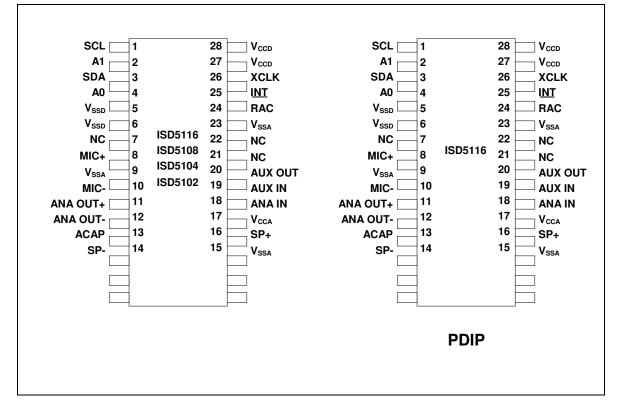
ISD5100

## 3. BLOCK DIAGRAM



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## 4. PIN CONFIGURATION



## 5. PIN DESCRIPTION

Pin Name	SOIC/PDIP	Functionality
SCL	1	I <sup>2</sup> C Serial Clock Line: to clock the data into and out of the I <sup>2</sup> C interface.
A1	2	Input pin that supplies the LSB +1 bit for the I <sup>2</sup> C Slave Address.
SDA	3	I <sup>2</sup> C Serial Data Line: Data is passed between devices on the bus over this line.
A0	4	Input pin that supplies the LSB for the I <sup>2</sup> C Slave Address.
V <sub>SSD</sub>	5,6	Digital Ground.
NC	7,21,22	No Connect.
MIC+	8	Differential Positive Input for the microphone amplifier.
$V_{SSA}$	9,15,23	Analog Ground.
MIC-	10	Differential Negative Input for the microphone amplifier.
ANA OUT+	11	Differential Positive Analog Output for ANA OUT.
ANA OUT-	12	Differential Negative Analog Output for ANA OUT.
ACAP	13	AGC/AutoMute Capacitor: Required for the on-chip AGC amplifier during record and AutoMute function during playback.
SP-	14	Differential Negative Speaker Output: When the speaker outputs are in use, the AUX OUT output is disabled.
SP+	16	Differential Positive Speaker Output.
V <sub>CCA</sub>	17	Analog Supply Voltage: This pin supplies power to the analog sections of the device. It should be carefully bypassed to Analog Ground to insure correct device operation.
ANA IN	18	Analog Input: one of the analog inputs with selectable gain.
AUX IN	19	Auxiliary Input: one of the analog inputs with selectable gain.
AUX OUT	20	Auxiliary Output: one the analog outputs of the device. When this output is used, the SP+ and SP- outputs are disabled.
RAC	24	Row Address Clock; an open drain output. The RAC pin goes LOW $T_{RACL}^{[1]}$ before the end of each row of memory and returns HIGH at exactly the end of each row of memory.
INT	25	Interrupt Output; an open drain output that indicates that a set EOM bit has been found during Playback or that the chip is in an Overflow (OVF) condition. This pin remains LOW until a Read Status command is executed.
XCLK	26	This pin must be grounded for utilizing internal clock. For precision timing control, external clock signal can be applied through this pin.
Vccd	27,28	Digital Supply Voltage. These pins supply power to the digital sections of the device. They must be carefully bypassed to Digital Ground to insure correct device operation.

<sup>&</sup>lt;sup>[1]</sup> See the <u>Parameters section</u>

## 6. FUNCTIONAL DESCRIPTION

#### 6.1. OVERVIEW

6.4

## 6.1.1 Speech/Voice Quality

The ISD5100 ChipCorder Series can be configured via software to operate at 4.0, 5.3, 6.4 or 8.0 kHz sampling frequency to select appropriate voice quality. Increasing the duration decreases the sampling frequency and bandwidth, which affects audio quality. The table in the following section shows the relationship between sampling frequency, duration and filter pass band.

## 6.1.2 Duration

 
 Sample Rate (kHz)
 Duration <sup>[1]</sup>
 Typical Filter Knee (kHz)

 8.0
 8 min 44 sec
 4 min 22 sec
 2 min 11 sec
 1 min 5 sec
 3.4

2 min 43 sec

1 min 21 sec

2.7

2.3

To meet system requirements, the ISD5100 Series are single-chip solution, which provide 1 to 16 minutes of voice record and playback, depending upon the sample rates chosen.

5.3	13 min 6 sec	6 min 33 sec	3 min 17 sec	1 min 38 sec
4.0	17 min 28 sec	8 min 44 sec	4 min 22 sec	2 min 11 sec

5 min 27 sec

<sup>[1]</sup> Minus any pages selected for digital storage

10 min 55 sec

## 6.1.3 Flash Technology

One of the benefits of Nuvoton's ChipCorder technology is the use of on-chip Flash 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 10,000 times (typically) for the digital data and over 100,000 times (typically) for the analog messages.

A new feature has been added that allows memory space in the ISD5100 Series to be allocated to either digital or analog storage when recorded. The fact that a section has been assigned digital or analog data is stored in the Message Address Table by the system microcontroller when the recording is made.

## 6.1.4 Microcontroller Interface

The ISD5100 Series are controlled through an I<sup>2</sup>C 2-wire interface. This synchronous serial port allows commands, configurations, address data, and digital data to be loaded into the device, while allowing status, digital data and current address information to be read back from the device. In addition to the serial interface, two other status pins can feedback to the microcontroller for enhanced interface. These

are the RAC timing pin and the INT pin for interrupts to the controller. Communications with all the internal registers of any operations are through the serial bus, as well as digital memory Read and Write operations.

## 6.1.5 Programming

The ISD5100 Series are also ideal for playback-only applications, where single or multiple messages may be played back when desired. Playback is controlled through the I<sup>2</sup>C interface. Once the desired

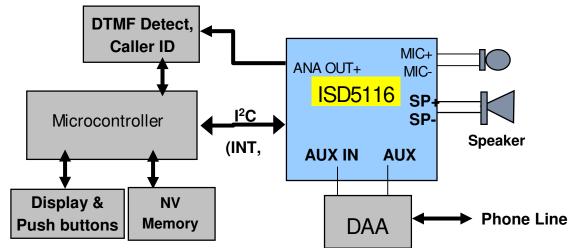
message configuration is created, duplicates can easily be generated via a third-party programmer. For more information on available application tools and programmers, please see the Nuvoton web site at http://www.nuvoton.com

#### 6.2. FUNCTIONAL DETAILS

The ISD5100 Series are single chip solutions for analog and digital data storage. The array can be divided between analog and digital storage according to user's choice, when the device is configured.

The below block diagram shows that the ISD5116 device can be easily designed into a telephone answering machine (TAD). Both Mic inputs transmit the voice input signal from the microphone to perform OGM recording, as well as to record the speech during phone conversation (simplex).

When the TAD is activated, the voice of the other party from the phone line feeds into the AUX IN, and is recorded into the ISD5116 device. Then the new message is usually indicated with blinking new message LED. Hence, during playback, the recorded message is sent out to speaker with volume control. Two I<sup>2</sup>C pins are used for all communications between the ChipCorder and the microcontroller for analog and/or digital storage, and the two outputs, INT and RAC are feedback to microcontroller for message management.



For duplex recording, speech from Mic inputs and message from received path can be directly recorded into the array simultaneously, then playback afterwards. In addition, for speaker phone operation, voice from Mic inputs are fed to AUX OUT and transmitted to the phone line, while message from other party is input from the AUX IN, then fed through to the speaker for listening.

The ISD5100 device has the flexibility for other applications, because the audio paths can be configured differently, with each circuit block being powered-up or –down individually, according to the applications requirement.

## 6.2.1 Internal Registers

The ISD5100 Series have multiple internal registers that are used to store the address information and the configuration or set-up of the device. The two 16-bit configuration registers control the audio paths through the device, the sample frequency, the various gains and attenuations, power up and down of different sections, and the volume settings. These registers are discussed in detail in <u>section 7.3.5</u>.

## 6.2.2 Memory Architecture

The ISD5100 Series memory array are arranged in various pages (or rows) of each 2048 bits as follows. The primary addressing for the pages are handled by 11 bits of address input in the analog mode.

A memory page is 2048 bits organized as thirty-two 64-bit "blocks" when used for digital storage. The contents of a page are either analog or digital. This is determined by instruction (opcode) at the time the data is written. A record of where is analog and where is digital, is stored in a message address table (MAT) by the system microcontroller. The MAT is a table kept in the microcontroller memory that defines the status of each message "page". It can be stored back into the ISD5100 Series if the power fails or the system is turned off. Using this table allows efficient message management. Segments of messages can be stored wherever there is available space in the memory array. [This is explained in detail for the ISD5008 in Applications Note #9 and will be similarly described in a later Note for the ISD5100-Series.]

Products	Pages (Rows)	Bits/Page	Memory Size
ISD5116	2048	2048	4,194,304 bits
ISD5108	1024	2048	2,097,152 bits
ISD5104	512	2048	1,048,576 bits
ISD5102	256	2048	524,288 bits

When a page is used for analog storage, the same 32 blocks are present but there are 8 EOM (End-of-Message) markers. This means that for each 4 blocks there is an EOM marker at the end. Thus, when recording, the analog recording will stop at any one of eight positions. At 8 kHz sampling frequency, this results in a resolution of 32 msec when ENDING an analog recording. Beginning an analog recording is limited to the 256 msec resolution provided by the 11-bit address. A recording does not immediately stop when the Stop command is given, but continues until the 32 millisecond block is filled. Then a bit is placed in the EOM memory to develop the interrupt that signals a message is finished playing in the Playback mode.

Digital data is sent and received serially over the I<sup>2</sup>C interface. The data is serial-to-parallel converted and stored in one of two alternating (commutating) 64-bit shift registers. When an input register is full, it becomes the register that is parallel written into the array. The prior write register becomes the new serial input register. A mechanism is built-in to ensure there is always a register available for storing new data.

Storing data in the memory is accomplished by accepting data one byte at a time and issuing an acknowledge. If data is coming in faster than it can be written, the chip issues an acknowledge to the host microcontroller, but holds SCL LOW until it is ready to accept more data. (See section 7.5.2 for details).

The read mode is the opposite of the write mode. Data is read into one of two 64-bit registers from the array and serially sent to the  $I^2C$  interface. (See section 7.5.3 for details).

#### **6.3. OPERATIONAL MODES DESCRIPTION**

### 6.3.1 I<sup>2</sup>C Interface

To use more than four ISD5100 Series devices in an application requires some external switching of the I<sup>2</sup>C interface.

#### I<sup>2</sup>C interface

**Important note:** The rest of this data sheet will assume that the reader is familiar with the I<sup>2</sup>C serial interface. Additional information on I<sup>2</sup>C may be found in <u>section 10</u> on page 72 of this document. If you are not familiar with this serial protocol, please read this section to familiarize yourself with it. A large amount of additional information on I<sup>2</sup>C can also be found on the Philips web page at <a href="http://www.philips.com/">http://www.philips.com/</a>.

#### I<sup>2</sup>C Slave Address

The ISD5100 Series have 7-bit slave address of <100 00xy> where x and y are equal to the state, respectively, of the external address pins A1 and A0. Because all data bytes are required to be 8 bits, the LSB of the address byte is the Read/Write selection bit that tells the slave whether to transmit or receive data. Therefore, there are 8 possible slave addresses for the ISD5100-Series. These are:

Fillout Table									
A0	Slave Address	R/W Bit	HEX Value						
0	<100 0000>	0	80						
1	<100 0001>	0	82						
0	<100 0010>	0	84						
1	<100 0011>	0	86						
0	<100 0000>	1	81						
1	<100 0001>	1	83						
0	<100 0010>	1	85						
1	<100 0011>	1	87						
	0 1 0 1 0 1	0       <100 0000>         1       <100 0001>         0       <100 0010>         1       <100 0011>         0       <100 0000>         1       <100 0000>         1       <100 0001>         0       <100 0001>         0       <100 0001>         0       <100 0001>         0       <100 0010>	0         <100 0000>         0           1         <100 0001>         0           0         <100 0010>         0           1         <100 0011>         0           1         <100 0001>         1           0         <100 0001>         1           1         <100 0000>         1           1         <100 0001>         1           0         <100 0001>         1						

**Pinout Table** 

## ISD5100 Series I<sup>2</sup>C Operation Definitions

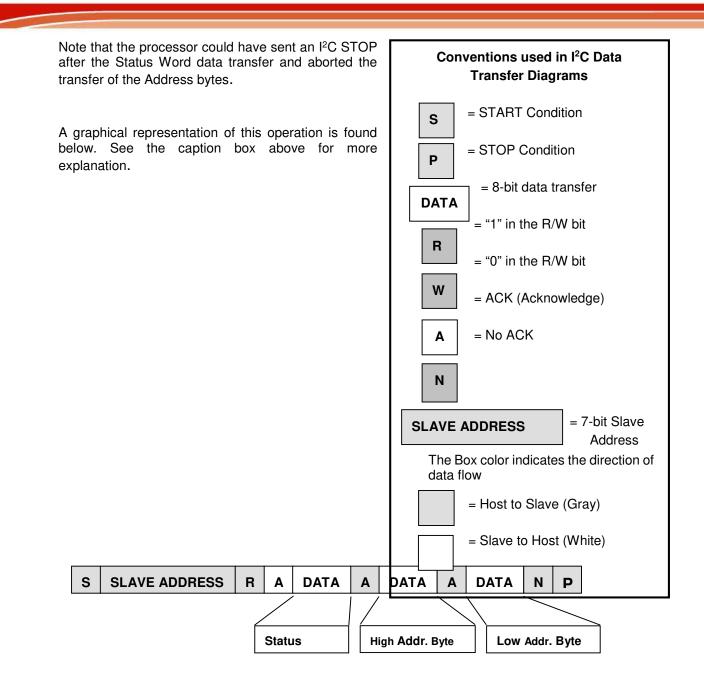
There are many control functions used to operate the ISD5100-Series. Among them are:

## 6.3.1.1. Read Status Command:

The Read Status command is a read request from the Host processor to the ISD5100 Series without delivering a Command Byte. The Host supplies all the clocks (SCL). In each case, the entity sending the data drives the data line (SDA). The Read Status Command is executed by the following I<sup>2</sup>C sequence.

- 1. Host executes I<sup>2</sup>C START
- 2. Send Slave Address with R/W bit = "1" (Read) 81h
- 3. Slave (ISD5100-Series) responds back to Host an Acknowledge (ACK) followed by 8-bit Status word
- 4. Host sends an Acknowledge (ACK) to Slave
- 5. Wait for SCL to go HIGH
- 6. Slave responds with Upper Address byte of internal address register
- 7. Host sends an ACK to Slave
- 8. Wait for SCL to go HIGH
- 9. Slave responds with Lower Address byte of internal address register (A[4:0] will always return set to 0.)
- 10. Host sends a NO ACK to Slave, then executes I<sup>2</sup>C STOP

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## 6.3.1.2. Load Command Byte Register (Single Byte Load):

A single byte may be written to the Command Byte Register in order to power up the device, start or stop Analog Record (if no address information is needed), or do a Message Cueing function. The Command Byte Register is loaded as follows:

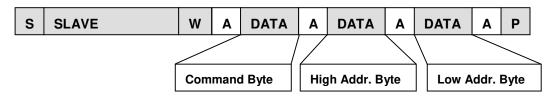


- 6. Slave responds with an ACK
- 7. Wait for SCL to go HIGH
- 8. Host executes I<sup>2</sup>C STOP

## 6.3.1.3. Load Command Byte Register (Address Load)

For the normal addressed mode the Registers are loaded as follows:

- 1. Host executes I<sup>2</sup>C START
- 2. Send Slave Address with R/W bit = "0" (Write)
- 3. Slave responds back with an ACK.
- 4. Wait for SCL to go HIGH
- 5. Host sends a byte to Slave (Command Byte)
- 6. Slave responds with an ACK
- 7. Wait for SCL to go HIGH
- 8. Host sends a byte to Slave (High Address Byte)
- 9. Slave responds with an ACK
- 10. Wait for SCL to go HIGH
- 11. Host sends a byte to Slave (Low Address Byte)
- 12. Slave responds with an ACK
- 13. Wait for SCL to go HIGH
- 14. Host executes I<sup>2</sup>C STOP



## 6.3.2 I<sup>2</sup>C Control Registers

The ISD5100 Series are controlled by loading commands to, or, reading from, the internal command, configuration and address registers. The Command byte sent is used to start and stop recording, write or read digital data and perform other functions necessary for the operation of the device.

#### Command Byte

Control of the ISD5100 Series are implemented through an 8-bit command byte, sent after the 7-bit device address and the 1-bit Read/Write selection bit. The 8 bits are:

- Global power up bit
- DAB bit: determines whether device is performing an analog or digital function
- 3 function bits: these determine which function the device is to perform in conjunction with the DAB bit.
- 3 register address bits: these determine if and when data is to be loaded to a register

Power Up Bit	C7	C6	C5	C4	C3	C2	C1	C0
• •	PU	DAB	FN2	FN1	FN0	RG2	RG1	RG0
			Functior	n Bits	Register Bits			

## **Function Bits**

The command byte function bits are detailed in the table to the right. C6, the DAB bit, determines whether the device is performing an analog or digital function. The other bits are decoded to produce the individual commands. Not all decode combinations are currently used, and are reserved for future use. Out of 16 possible codes, the ISD5100 Series uses 7 for normal operation. The other 9 are undefined

	Functio	on Bits		Function
C6	C5	C4	C3	
DAB	FN2	FN1	FN0	
0	0	0	0	STOP (or do nothing)
0	1	0	1	Analog Play
0	0	1	0	Analog Record
0	1	1	1	Analog MC
1	1	0	0	Digital Read
1	0	0	1	Digital Write
1	0	1	0	Erase (row)

## **Register Bits**

The register load may be used to modify a command sequence (such as load an address) or used with the null command sequence to load a configuration or test register. Not all registers are accessible to the user. [RG2 is always 0 as the four additional combinations are undefined.]

RG2	RG1	RG0	Function
C2	C1	C0	
0	0	0	No action
0	0	1	Reserved
0	1	0	Load CFG0
0	1	1	Load CFG1

## 6.3.3 Opcode Summary

#### **OpCode Command Description**

The following commands are used to access the chip through the I<sup>2</sup>C interface.

- Play: analog play command
- Record: analog record command
- Message Cue: analog message cue command
- Read: digital read command
- Write: digital write command
- Erase: digital page and block erase command
- Power up: global power up/down bit. (C7)
- Load CFG0: load configuration register 0
- Load CFG1: load configuration register 1
- Read STATUS: Read the interrupt status and address register, including a hardwired device ID

## **OPCODE COMMAND BYTE TABLE**

		Pwr		Functio	on Bits		Re	gister E	Bits
OPCODE	HEX	PU	DAB	FN2	FN1	FN0	RG2	RG1	RG0
COMMAND BIT NUMBER	CMD	C7	C6	C5	C4	C3	C2	C1	C0
POWER UP	80	1	0	0	0	0	0	0	0
POWER DOWN	00	0	0	0	0	0	0	0	0
STOP (DO NOTHING) STAY ON	80	1	0	0	0	0	0	0	0
STOP (DO NOTHING) STAY OFF	00	0	0	0	0	0	0	0	0
LOAD CFG0	82	1	0	0	0	0	0	1	0
LOAD CFG1	83	1	0	0	0	0	0	1	1
RECORD ANALOG	90	1	0	0	1	0	0	0	0
RECORD ANALOG @ ADDR	91	1	0	0	1	0	0	0	1
PLAY ANALOG	A8	1	0	1	0	1	0	0	0
PLAY ANALOG @ ADDR	A9	1	0	1	0	1	0	0	1
MSG CUE ANALOG	B8	1	0	1	1	1	0	0	0
MSG CUE ANALOG @ ADDR	B9	1	0	1	1	1	0	0	1
ENTER DIGITAL MODE	C0	1	1	0	0	0	0	0	0
EXIT DIGITAL MODE	40	0	1	0	0	0	0	0	0
DIGITAL ERASE PAGE	D0	1	1	0	1	0	0	0	0
DIGITAL ERASE PAGE @ ADDR	D1	1	1	0	1	0	0	0	1
DIGITAL WRITE	C8	1	1	0	0	1	0	0	0
DIGITAL WRITE @ ADDR	C9	1	1	0	0	1	0	0	1
DIGITAL READ	E0	1	1	1	0	0	0	0	0
DIGITAL READ @ ADDR	E1	1	1	1	0	0	0	0	1
READ STATUS <sup>1</sup>	N/A	N/A	N/A	N/A	N/A	N/A	N/A	N/A	N/A
1 Conception 7.0 on n									

1. See <u>section 7.2</u> on page 12 for details.

## 6.3.4 Data Bytes

In the I<sup>2</sup>C write mode, the device can accept data sent after the command byte. If a register load option is selected, the next two bytes are loaded into the selected register. The format of the data is MSB first, the I<sup>2</sup>C standard. Thus to load DATA<15:0> into the device, DATA<15:8> is sent first, the byte is acknowledged, and DATA<7:0> is sent next. The address register consists of two bytes. The format of the address is as follows:

#### ADDRESS<15:0> = PAGE\_ADDRESS<10:0>, BLOCK\_ADDRESS<4:0>

## Note: if an analog function is selected, the block address bits must be set to 00000. Digital Read and Write are block addressable.

When the device is polled with the Read Status command, it will return three bytes of data. The first byte is the status byte, the next the upper address byte and the last the lower address byte. The status register is one byte long and its bit function is:

#### STATUS<7:0> = EOM, OVF, READY, PD, PRB, DEVICE\_ID<2:0>

Lower address byte will always return the block address bits as zero, either in digital or analog mode.

EOM	BIT 7	Indicates whether an EOM interrupt has occurred.					
OVF	BIT 6	Indicates whether an overflow interrupt has occurred.					
READY	BIT 5	Indicates the internal status of the device – if READY is LOW no new commands should be sent to device, i.e. Not Ready.					
PD	BIT 4	Device is powered down if PD is HIGH.					
PRB	BIT 3	Play/Record mode indicator. HIGH=Play/LOW=Record.					
DEVICE_ID	BIT 0, 1, 2	An internal device ID. ISD5116 = 001; ISD5108 = 010; ISD5104 = 100 and ISD5102 = 101.					

The functions of the bits are:

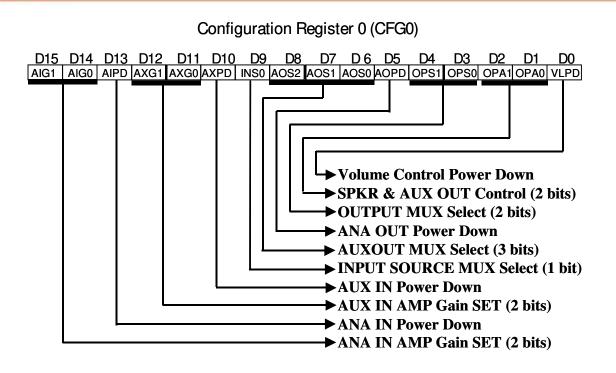
It is recommended that you read the status register after a Write or Record operation to ensure that the device is ready to accept new commands. Depending upon the design and the number of pins available on the controller, the polling overhead can be reduced. If INT and RAC are tied to the

## microcontroller, it does not have to poll as frequently to determine the status of the ISD5100-SERIES.

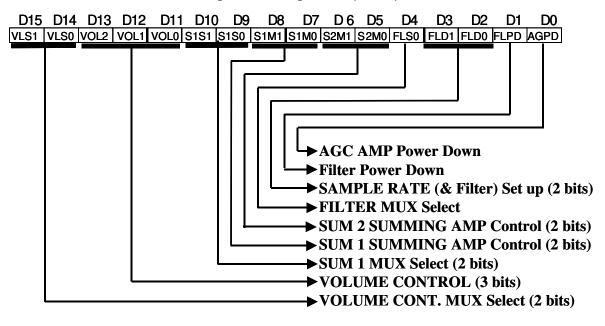
## 6.3.5 Configuration Register Bytes

The configuration register bytes are defined, in detail, in the drawings of <u>section 7.4</u> on page 29. The drawings display how each bit enables or disables a function of the audio paths in the ISD5100-Series. The tables below give a general illustration of the bits. There are two configuration registers, CFG0 and CFG1, so there are four 8-bit bytes to be loaded during the set-up of the device.





Configuration Register 1 (CFG1)



## 6.3.6 Power-up Sequence

This sequence prepares the ISD5100 Series for an operation to follow, waiting the Tpud time before sending the next command sequence.

- 1. Send I<sup>2</sup>C POWER UP
- 2. Send one byte 10000000 {Slave Address, R/W = 0} 80h
- 3. Slave ACK
- 4. Wait for SCL High
- 5. Send one byte 10000000 {Command Byte = Power Up} 80h
- 6. Slave ACK
- 7. Wait for SCL High
- 8. Send I<sup>2</sup>C STOP

#### Playback Mode

The command sequence for an analog Playback operation can be handled several ways. The most straightforward approach would be to incorporate a single four byte exchange, which consists of the Slave Address (80h), the Command Byte (A9h) for Play Analog @ Address, and the two address bytes.

#### Record Mode

The command sequence for an Analog Record would be a four byte sequence consisting of the Slave Address (80h), the Command Byte (91h) for Record Analog @ Address, and the two address bytes. See <u>"Load Command Byte Register (Address Load)" in section 7.3.2</u> on page 17.

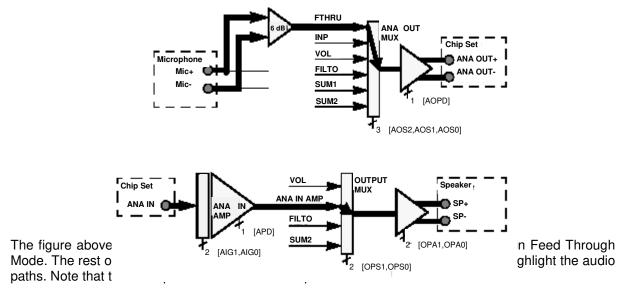
## 6.3.7 Feed Through Mode

The previous examples were dependent upon the device already being powered up and the various paths being set through the device for the desired operation. To set up the device for the various paths requires loading the two 16-bit Configuration Registers with the correct data. For example, in the Feed Through Mode the device only needs to be powered up and a few paths selected.

This mode enables the ISD5100 Series to connect to a cellular or cordless base band 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 Nuvoton 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 Nuvoton chip. This allows the Nuvoton chip to substitute for those functions and incidentally gain access to the audio to and from the base band chip set.

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To set up the environment described above, a series of commands need to be sent to the ISD5100-Series. First, the chip needs to be powered up as described in this section. Then the Configuration Registers must be filled with the specific data to connect the paths desired. In the case of the Feed Through Mode, most of the chip can remain powered down. The following figure illustrates the affected paths.



To select this mode, the following control bits must be configured in the ISD5100 Series 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 ANA OUT MUX. These are the D6, D7 and D8 bits respectively of Configuration 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:

- 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. The <u>ANA IN Amplifier Gain Settings table</u> on page 36 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 9 dB 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 its 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 ISD5100 Series chip must be defined before the configuration registers settings are updated:

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- 1. Power down the Volume Control Element—Bit VLPD controls the power up state of the Volume Control. This is bit D0 of CFG0 and it should be set to a ONE to power down this stage.
- 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 this stage.
- 3. Power down the SUM1 and SUM2 Mixer amplifiers—Bits S1M0 and S1M1 control the SUM1 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 these two amplifiers.
- 4. Power down the FILTER stage—Bit FLPD controls the power up state of the FILTER stage in the device. This is bit D1 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 this 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 a ZERO. (a). Bit INSO, 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 440B)

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 SCL.

## 6.3.8 Call Record

The call record mode adds the ability to record an incoming phone call. In most applications, the ISD5100 Series would first be set up for Feed Through Mode as described above. When the user wishes to record the incoming call, the setup 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 ISD5100 Series 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.
- 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 must be set to ZERO to select the SUM1 SUMMING amplifier path.
- 4. 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 must 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 must be set to ONE and D3 set to ZERO.
- 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.

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.

## 6.3.9 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 must be set to ZERO to power up this stage.
- 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 must be set to a ZERO to select the AGC amplifier.
- Select the INPUT SOURCE MUX (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 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 must 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 must 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 must be set to ZERO and D3 set to ONE.
- 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.

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.

## 6.3.10 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 piezo 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, the state of the FILTER MUX. This is bit D4 of CFG1 and must be set to ONE to select the MULTILEVEL STORAGE ARRAY.
- 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 must 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.
- Select the SUM2 SUMMING amplifier path through the VOLUME MUX—Bits VLS0 and VLS1 control the state VOLUME MUX. These bits are bits 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 must be set to a ZERO to power-up the VOLUME CONTROL.
- 7. Select a VOLUME CONTROL LEVEL—Bits VOL0, VOL1, and VOL2 control the state of the VOLUME CONTROL LEVEL. These are bits D11, D12, and D13, respectively, of CFG1. A binary count of 000 through 111 controls the amount of attenuation through 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 should be set to a ZERO.
- 8. Select the VOLUME CONTROL path through the OUTPUT MUX—These are bits D3 and D4, respectively, of CFG0. They should be set to the state where D3 is ZERO and D4 is a 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 must 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 0100 (hex 2424).

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

Only those portions necessary for this mode are powered up.

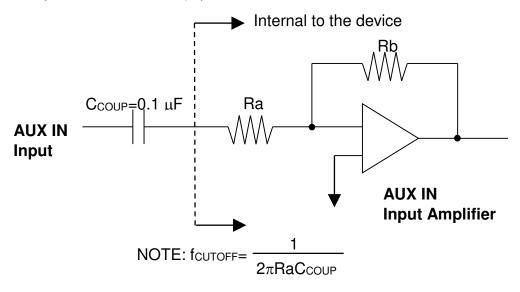
## 6.3.11 Message Cueing

Message cueing allows the user to skip through analog messages without knowing the actual physical location of the message. This operation is used during playback. In this mode, the messages are skipped 512 times faster than in normal playback mode. It will stop when an EOM marker is reached. Then, the internal address counter will be pointing to the next message.

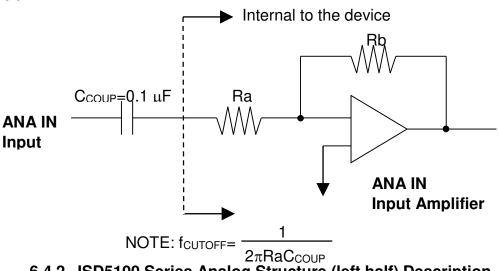
#### 6.4. ANALOG MODE

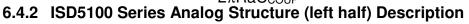
### 6.4.1 Aux In and Ana In Description

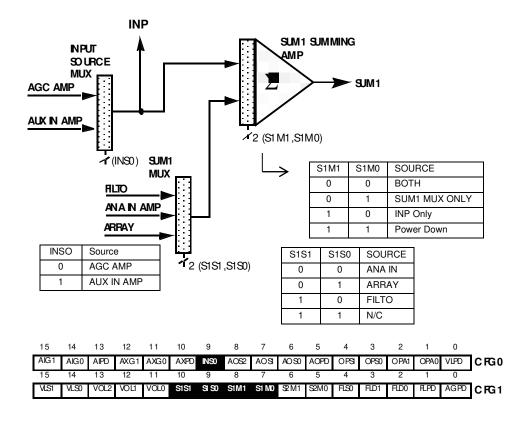
The AUX IN is an additional audio input to the ISD5100-Series, such as from the microphone circuit in a mobile phone "car kit." This input has a nominal 694 mV p-p level at its minimum gain setting (0 dB). See the <u>AUX IN Amplifier Gain Settings table</u> on page 37. Additional gain is available in 3 dB steps (controlled by the I<sup>2</sup>C serial interface) up to 9 dB.



The ANA IN pin is the analog input from the telephone chip set. It can be switched (by the serial bus) to the speaker output, the array input or to various other paths. This pin is designed to accept a nominal 1.11 Vp-p when at its minimum gain (6 dB) setting. See the ANA IN Amplifier Gain Settings table on page 37. There is additional gain available in 3 dB steps controlled from the I<sup>2</sup>C interface, if required, up to 15 dB.

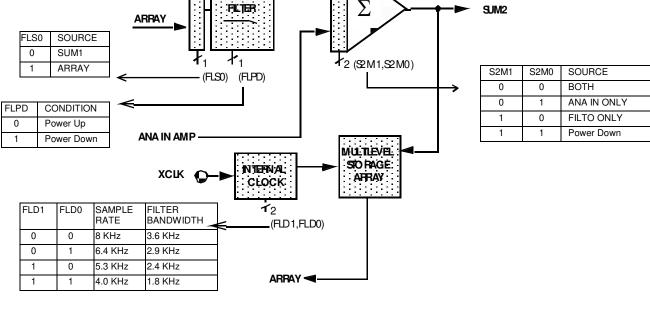






## 6.4.3 ISD5100 Series Aanalog Structure (right half) Description

#### FLS0 SOURCE



SUM2 SUMMING

AMP

6 5 2 15 14 13 12 11 10 9 8 7 4 3 1 0 VIST VISO VOL2 VOL1 VOL0 SISI SISO SIMI SIMO S2MI S2MO FLSO FLDI FLDO FLPD AGPD CFG1

FILTO

LOW PASS

FILTER

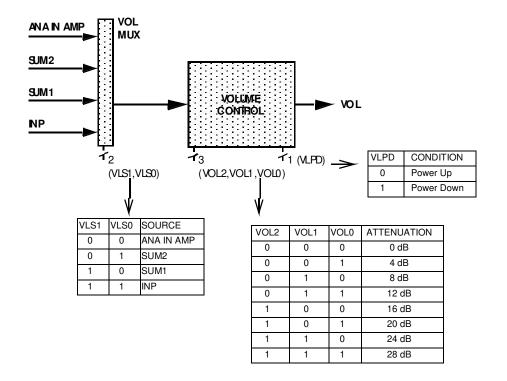
MUX

SUM1

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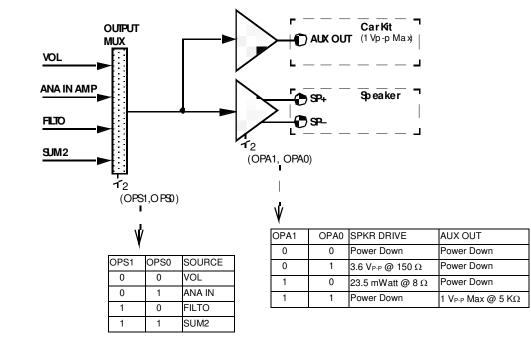
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## 6.4.4 Volume Control Description



AIG1	AIG0	AIPD	AXG1	AXG0	AXPD	INS0	AOS2	AO S1	AOS0	AOPD	OPS1	OPS0	OPA1	O PAO	VLPD	CFG0
15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	-
VISI	VLSO	VOI2	VOL1	VOLO	SI SI	S1S0	S1M1	SI MO	S2M1	S2M0	FLSD	FLD1	FLD0	FLPD	AGPD	CFG1

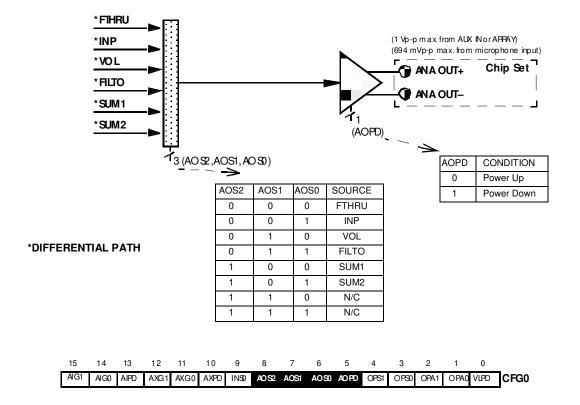
## 6.4.5 Speaker and Aux Out Description



15	14	13	12	11	10	9	8	7	6	5	4	3	2	1	0	
AIG1	AIG0	APD	AXG1	AXG0	AXPD	INS0	AOS2	AO S1	AOS0	AOPD	OPSI	0PS0	OPA1	O PAO	VLPD	CFG0

## 6.4.6 Ana Out Description

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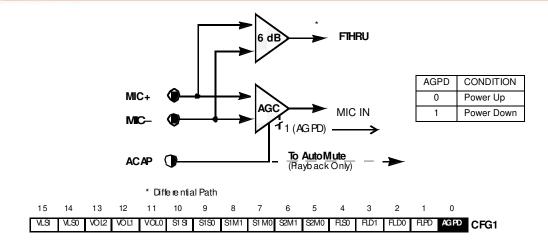
## 6.4.7 Analog Inputs

#### Microphone Inputs

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 direct path to the ANA OUT MUX has a gain of 6 dB so a 208 mV p-p signal across the differential microphone inputs would give 416 mV p-p across the ANA OUT pins. The AGC circuit has a range of 45 dB in order to deliver a nominal 694 mV p-p into the storage array from a typical electric microphone output of 2 to 20 mV p-p. The input impedance is typically 10k $\Omega$ .

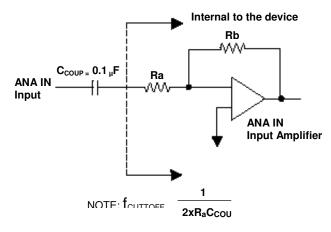
The ACAP pin provides the capacitor connection for setting the parameters of the microphone AGC circuit. It should have a 4.7  $\mu$ F capacitor connected to ground. It cannot be left floating. This is because the capacitor is also used in the playback mode for the AutoMute circuit. This circuit reduces the amount of noise present in the output during quiet pauses. Tying this pin to ground gives maximum gain; to VCCA gives minimum gain for the AGC amplifier but will cancel the AutoMute function.





#### ANA IN (Analog Input)

The ANA IN pin is the analog input from the telephone chip set. It can be switched (by the I<sup>2</sup>C interface) to the speaker output, the array input or to various other paths. This pin is designed to accept a nominal 1.11 V p-p when at its minimum gain (6 dB) setting. There is additional gain available, if required, in 3 dB steps, up to 15 dB. The gain settings are controlled from the I<sup>2</sup>C interface.



Gain Setting	Resistor Ratio (Rb/Ra)	Gain	Gain² (dB)
00	63.9 / 102	0.625	-4.1
01	77.9 / 88.1	0.883	-1.1
10	92.3 / 73.8	1.250	1.9
11	106 / 60	1.767	4.9

Note: Ra & Rb are in  $k\Omega$ 

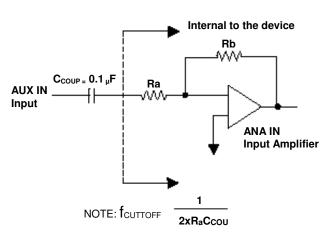
## **ANA IN Amplifier Gain Settings**

Setting <sup>(1)</sup>	0TLP Input	CF	G0	Gain <sup>(2)</sup>	Array	Speaker
	$V_{P-P}^{(3)}$	AIG1	AIG0		In/Out V <sub>P-P</sub>	Out V <sub>P-P</sub> <sup>(4)</sup>
6 dB	1.110	0	0	0.625	0.694	2.22
9 dB	0.785	0	1	0.883	0.694	2.22
12 dB	0.555	1	0	1.250	0.694	2.22
15 dB	0.393	1	1	1.767	0.694	2.22

4. Speaker Out gain set to 1.6 (High). (Differential)

## AUX IN (Auxiliary Input)

The AUX IN is an additional audio input to the ISD5100-Series, such as from the microphone circuit in a mobile phone "car kit." This input has a nominal 694 mV p-p level at its minimum gain setting (0 dB). See the following table. Additional gain is available in 3 dB steps (controlled by the I<sup>2</sup>C interface) up to 9 dB.



## **AUX IN Input Modes**

Gain Setting	Resistor Ratio (Rb/Ra)	Gain	Gain <sup>(2)</sup> (dB)
00	40.1 / 40.1	1.0	0
01	47.0 / 33.2	1.414	3
10	53.5 / 26.7	2.0	6
11	59.2 / 21	2.82	9

Note: Ra & Rb are in  $k\Omega$ 

## **AUX IN Amplifier Gain Settings**

Setting <sup>(1)</sup>	0TLP Input	CFG0		Gain <sup>(2)</sup>	Array	Speaker
	$V_{P-P}^{(3)}$	AXG1	AXG0		In/Out V <sub>P-P</sub>	Out V <sub>P-P</sub> <sup>(4)</sup>
0 dB	0.694	0	0	1.00	0.694	0.694
3 dB	0.491	0	1	1.41	0.694	0.694
6 dB	0.347	1	0	2.00	0.694	0.694
9 dB	0.245	1	1	2.82	0.694	0.694

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

#### 6.5. DIGITAL MODE

## 6.5.1 Erasing Digital Data

The Digital Erase command can only erase an entire page at a time. This means that the D1 command only needs to include the 11-bit page address; the 5-bit for block address are left at 00000.

Once a page has been erased, each block may be written separately, 64 bits at a time. But, if a block has been previously written then the entire page of 2048 bits must be erased in order to re-write (or change) a block.

A sequence might be look like:

- read the entire page
- store it in RAM
- change the desired bit(s)
- erase the page
- write the new data from RAM to the entire page

## 6.5.2 Writing Digital Data

The Digital Write function allows the user to select a portion of the array to be used as digital memory. The partition between analog and digital memory is left up to the user. A page can only be either Digital or Analog, but not both. The minimum addressable block of memory in the digital mode is one block or 64 bits, when reading or writing. The address sent to the device is the 11-bit row (or page) address with the 5-bit scan (or block) address. However, one must send a Digital Erase before attempting to change digital data on a page. This means that even when changing only one of the 32 blocks, all 32 blocks will need to be rewritten to the page. Command Sequence: The chip enters digital mode by sending the ENTER DIGITAL MODE command from power down. Send the DIGITAL WRITE @ ADDR command with the row address. After the address is entered, the data is sent in one-byte packets followed by an I<sup>2</sup>C acknowledge generated by the chip. Data for each block is sent MSB first. The data transfer is ended when the master generates an I<sup>2</sup>C STOP condition. If only a partial block of data is sent before the STOP condition, "zero" is written in the remaining bytes; that is, they are left at the erase level. An erased page (row) will be read as all zeros. The device can buffer up to two blocks of data. If the device is unable to accept more data due to the internal write process, the SCL line will be held LOW indicating to the master to halt data transfer. If the device encounters an overflow condition, it will respond by generating an interrupt condition and an I<sup>2</sup>C Not Acknowledge signal after the last valid byte of data. Once data transfer is terminated, the device needs up to two cycles (64 us) to complete its internal write cycle before another command is sent. If an active command is sent before the internal cycle is finished, the part will hold SCL LOW until the current command is finished. After writing is complete, send the EXIT DIGITAL MODE command.

## 6.5.3 Reading Digital Data

The Digital Read command utilizes the combined I<sup>2</sup>C command format. That is, a command is sent to the chip using the write data direction. Then the data direction is reversed by sending a repeated start condition, and the slave address with R/W set to 1. After this, the slave device (ISD5100-Series) begins to send data to the master until the master generates a NACK. If the part encounters an overflow

condition, the INT pin is pulled LOW. No other communication with the master is possible due to the master generating ACK signals.

Digital Write and Digital Read can be done a "block" at a time. Thus, only 64 bits need be read in each Digital Read command sequence.

## 6.5.4 Example Command Sequences

An explanation and graphical representation of the Erase, Write and Read operations are found below.

Note: All sequences assumes that the chip is in power-down mode before the commands are sent.

#### 6.5.4.1. **Erase Digital Data**

Erase							
===== I2CStart SendByte(0x80) - Write, Slave address zero WaitACK WaitSCLHigh							
SendByte(0xc0) - Enter Digital Mode Command WaitACK WaitSCLHigh I2CStop							
I2CStart SendByte(0x80) - Write, Slave address zero WaitACK WaitSCLHigh							
SendByte(0xd1) - Digital Erase Command WaitACK WaitSCLHigh							
SendByte(row/256) - high address byte WaitACK WaitSCLHigh							
SendByte(row%256) - low address byte WaitACK WaitSCLHigh I2CStop							
repeat until the number of RAC pulses are one less than the number of rows to delete {							
wait RAC low WAIT RAC high }							
Note: If only one row is going to be erased, send the following STOP command immediately after ERASE command and skip the loop above							
I2CStart SendByte(0x80) - Write, Slave address zero WaitACK WaitSCLHigh							

WaitSCLHigh SendByte(0xc0) - Stop digital erase

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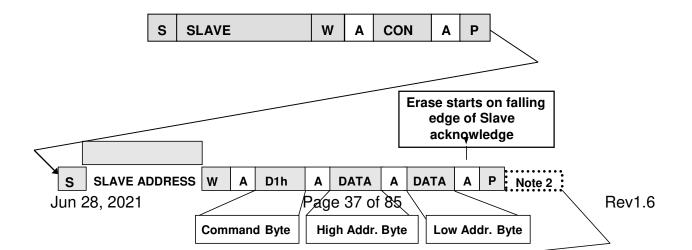
```
WaitACK
WaitSCLHigh
I2CStop
wait until erase of
```

```
wait until erase of the last row has completed
{
    wait RAC low
    WAIT RAC high
}
```

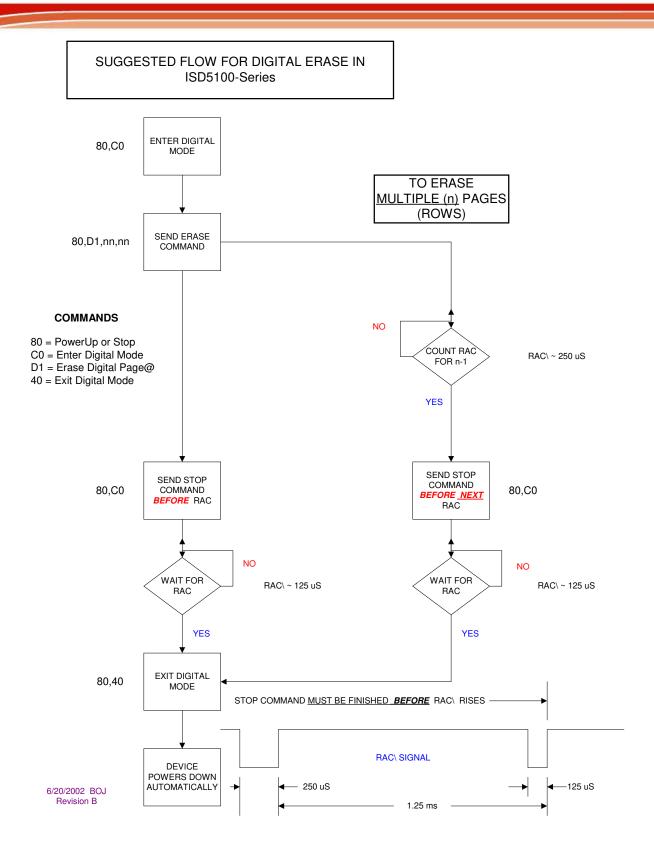
```
I2CStart
SendByte(0x80) - Write, Slave address zero
WaitACK
WaitSCLHigh
SendByte(0x40) - Exit Digital Mode Command
WaitACK
WaitSCLHigh
I2Cstop
```

#### Notes

- 1. Erase operations must be addressed on a Row boundary. The 5 LSB bits of the Low Address Byte will be ignored.
- 2. I<sup>2</sup>C bus is released while erase proceeds. Other devices may use the bus until it is time to execute the STOP command that causes the end of the Erase operation.
- 3. Host processor must count RAC cycles to determine where the chip is in the erase process, one row per RAC cycle. RAC pulses LOW for 0.25 millisecond at the end of each erased row. The erase of the "next" row begins with the rising edge of RAC. See the <u>Digital Erase RAC</u> timing diagram on page 51.
- 4. When the erase of the last desired row begins, the following STOP command (Command Byte = 80 hex) must be issued. This command must be completely given, including receiving the ACK from the Slave before the RAC pin goes HIGH at the end of the row.









#### Write Digital Data 6.5.4.2.

#### Write

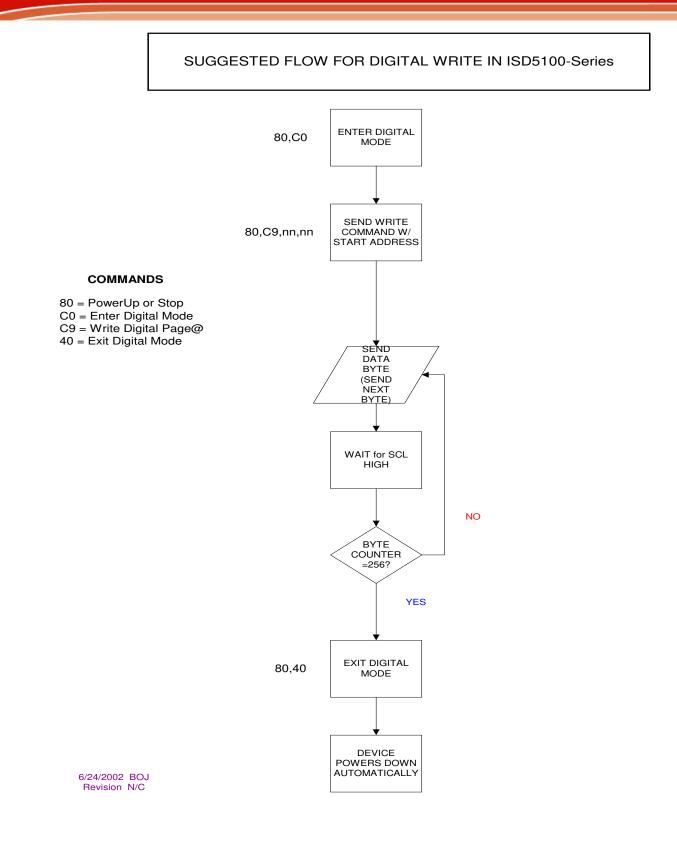
=====

I2CStart SendByte(0x80) WaitACK WaitSCLHigh SendByte(0xc0) WaitACK WaitSCLHigh I2CStop	- Write, Slave address zero - Enter Digital Mode Command
I2CStart SendByte(0x80) WaitACK	- Write, Slave address zero
WaitSCLHigh SendByte(0xc9) WaitACK WaitSCLHigh	- Write Digital Data Command
SendByte(row/256) WaitACK	- high address byte
WaitSCLHigh SendByte(row%256) WaitACK WaitSCLHigh	- low address byte
repeat until all data is	sent
{ SendByte(data) WaitACK() WaitSCLHigh()	- send data byte
} I2CStop	
I2CStart SendByte(0x80) WaitACK	- Write, Slave address zero
WaitSCLHigh SendByte(0x40) WaitACK WaitSCLHigh I2CStop	- Exit Digital Mode Command
S	SLAVE W A CON
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Δ Ρ





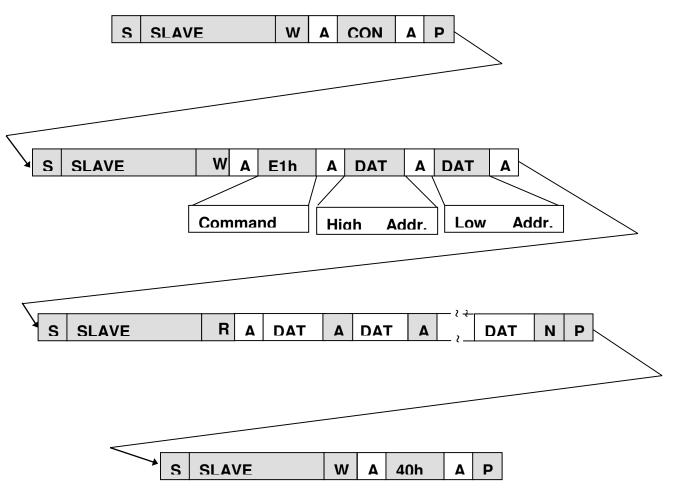


# 6.5.4.3. Read Digital Data

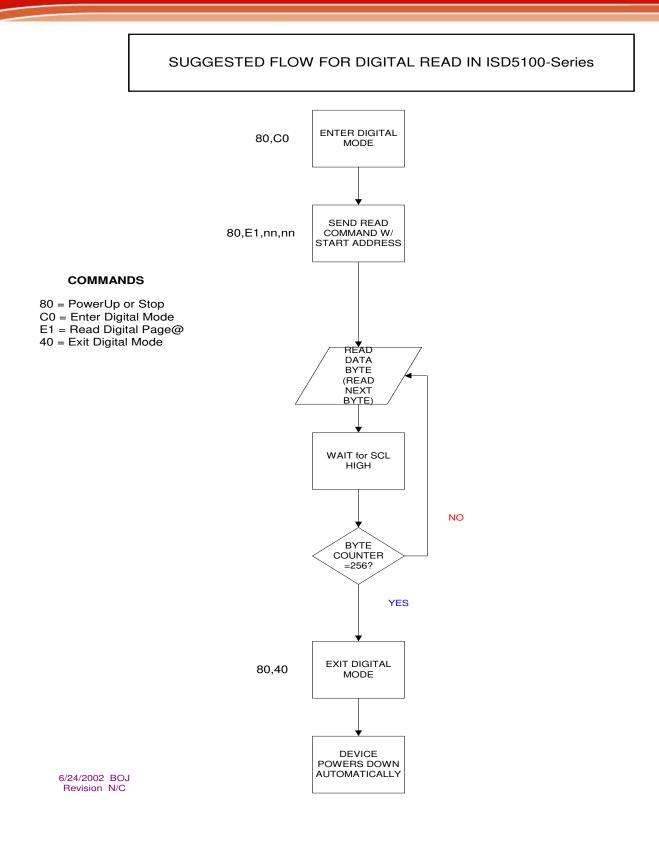
Read =====	
I2CStart SendByte(0x80) WaitACK	- Write, Slave address zero
WaitSCLHigh SendByte(0xc0) WaitACK WaitSCLHigh	- Enter Digital Mode
I2CStop	
I2CStart SendByte(0x80) WaitACK WaitSCLHigh	- Write, Slave address zero
SendByte(0xe1) WaitACK	- Read Digital Data Command
WaitSCLHigh SendByte(row/256) WaitACK	- high address byte
WaitSCLHigh() SendByte(row%256) WaitACK WaitSCLHigh	- low address byte
I2CStart SendByte(0x81)	<ul> <li>Send repeat start command</li> <li>Read, Slave address zero</li> </ul>
repeat until all data is	read
{ data = ReadByte SendACK WaitSCLHigh }	<ul> <li>() - send clocks to read data byte</li> <li>- send NACK on the last byte</li> <li>- The only flow control available</li> </ul>
, I2CStop()	
1200100()	
I2CStart SendByte(0x80) WaitACK WaitSCLHigh	- Write, Slave address zero
SendByte(0x40) WaitACK WaitSCLHigh	- Exit Digital Mode

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I2CStop







#### 6.6. PIN DETAILS

#### 6.6.1 Digital I/O Pins

#### SCL (Serial Clock Line)

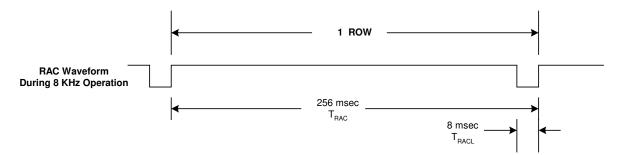
The Serial Clock Line is a bi-directional clock line. It is an open-drain line requiring a pull-up resistor to Vcc. It is driven by the "master" chips in a system and controls the timing of the data exchanged over the Serial Data Line.

#### SDA (Serial Data Line)

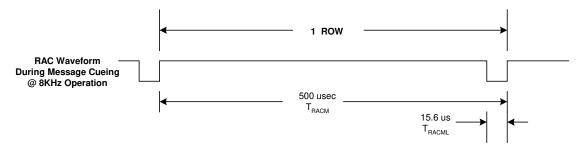
The Serial Data Line carries the data between devices on the I<sup>2</sup>C interface. Data must be valid on this line when the SCL is HIGH. State changes can only take place when the SCL is LOW. This is a bidirectional line requiring a pull-up resistor to Vcc.

#### RAC (Row Address Clock)

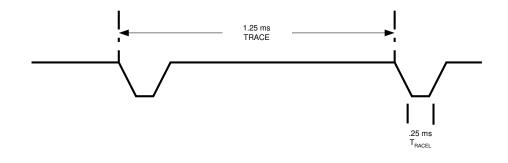
RAC is an open drain output pin that normally marks the end of a row. At the 8 kHz sample frequency, the duration of this period is 256 ms. RAC stays HIGH for 248 ms and stays LOW for the remaining 8 ms before it reaches the end of a row. There are 2048 rows of memory in the ISD5116 devices, 1024 rows in the ISD5108, 512 rows in the ISD5104 and 256 rows in the ISD5102.



The RAC pin remains HIGH for 500  $\mu$ sec and stays LOW for 15.6  $\mu$ sec under the Message Cueing mode. See the <u>Timing Parameters table</u> on page 64 for RAC timing information at other sample rates. When a record command is first initiated, the RAC pin remains HIGH for an extra T<sub>RACML</sub> period, to load sample and hold circuits internal to the device. The RAC pin can be used for message management techniques.



## RAC Waveform During Digital Erase @ 8kHz Operation



#### INT (Interrupt)

INT is an open drain output pin. The ISD5100 Series interrupt pin goes LOW and stays LOW when an Overflow (OVF) or End of Message (EOM) marker is detected. Each operation that ends in an EOM or OVF generates an interrupt, including the message cueing cycles. The interrupt is cleared by a READ STATUS instruction that will give a status byte out the SDA line.

#### XCLK (External Clock Input)

This is the external clock input. To use internal clock, this pin must be grounded (suggest connecting to  $V_{SSD}$ ). While in internal clock mode, the ISD5100 Series are operated at one of four internal rates selected for its internal oscillator by the Sample Rate Select bits. For precision timing control, external clock signal can be applied through this pin. In the external clock mode, the device can be clocked through the XCLK pin at 4.096 MHz as described in <u>section 7.4.3</u> on page 32.

Because the anti-aliasing and smoothing filters track the Sample Rate Select bits, one must, for optimum performance, maintain the external clock at 4.096 MHz *AND* set the Sample Rate Configuration bits to one of the four values to properly set the filters to the correct cutoff frequency as described in <u>section</u> 7.4.3 on page 32. The duty cycle on the input clock is not critical, as the clock is immediately divided by two internally.

ISD5116 Duration (Minutes)	ISD5108 Duration (Minutes)	ISD5104 Duration (Minutes)	ISD5102 Duration (Minutes)	Sample Rate (kHz)	Required Clock (kHz)	FLD1	FLD0	Filter Knee (kHz)
8.73	4.36	2.18	1.08	8.0	4096	0	0	3.4
10.9	5.45	2.72	1.35	6.4	4096	0	1	2.7
13.1	6.55	3.27	1.63	5.3	4096	1	0	2.3
17.5	8.75	4.37	2.18	4.0	4096	1	1	1.7

#### **External Clock Input Table**

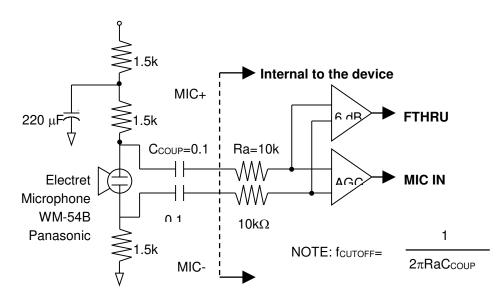
#### A0, A1 (Address Pins)

These two pins are normally strapped for the desired address that the ISD5100 Series will have on the  $I^2C$  serial interface. If there are four of these devices on the bus, then each must be strapped differently in order to allow the Master device to address them individually. The possible addresses range from 80h to 87h, depending upon whether the device is being written to, or read from, by the host. The ISD5100 Series have a 7-bit slave address of which only A0 and A1 are pin programmable. The eighth bit (LSB) is the R/W bit. Thus, the address will be 1000 0xy0 or 1000 0xy1. (See the table in section 7.3.1 on page 13.)

#### 6.6.2 Analog I/O Pins

#### MIC+, MIC- (Microphone Input +/-)

The microphone input transfers the voice signal to the on-chip AGC preamplifier or directly to the ANA OUT MUX, depending on the selected path. The direct path to the ANA OUT MUX has a gain of 6 dB so a 208 mV p-p signal across the differential microphone inputs would give 416 mV p-p across the ANA OUT pins. The AGC circuit has a range of 45 dB in order to deliver a nominal 694 mV p-p into the storage array from a typical electret microphone output of 2 to 20 mV p-p. The input impedance is typically 10k $\Omega$ .



#### ANA OUT+, ANA OUT- (Analog Output +/-)

This differential output is designed to go to the microphone input of the telephone chip set. It is designed to drive a minimum of 5 k $\Omega$  between the "+" and "-" pins to a nominal voltage level of 694 mV p-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.

#### ACAP (AGC Capacitor)

This pin provides the capacitor connection for setting the parameters of the microphone AGC circuit. It should have a 4.7  $\mu$ F capacitor connected to ground. It cannot be left 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; tying it to V<sub>CCA</sub> gives minimum gain for the AGC amplifier but cancels the AutoMute function.

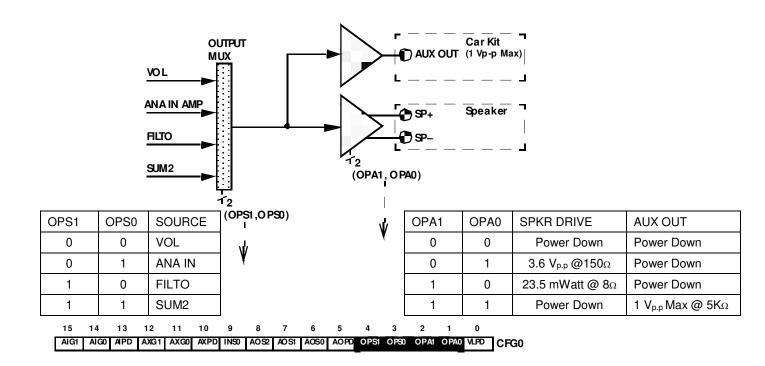
#### SP +, SP- (Speaker +/-)

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This is the speaker differential output circuit. It is designed to drive an  $8\Omega$  speaker connected across the speaker pins up to a maximum of 23.5 mW RMS power. 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.

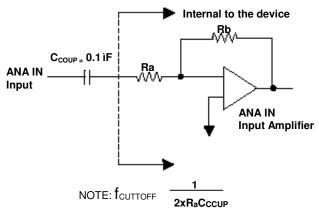
#### AUX OUT (Auxiliary Output)

The AUX OUT 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  $5k\Omega$  and up to a maximum of 1V p-p. The AC signal is superimposed on approximately 1.2 VDC bias and must be capacitively coupled to the load.



#### ANA IN (Analog Input)

The ANA IN pin is the analog input from the telephone chip set. It can be switched (by the I<sup>2</sup>C interface) to the speaker output, the array input or to various other paths. This pin is designed to accept a nominal 1.11 V p-p when at its minimum gain (6 dB) setting. There is additional gain available, if required, in 3 dB steps, up to 15 dB. The gain settings are controlled from the I<sup>2</sup>C interface.



# ANA IN Input Modes

Gain Setting	Resistor Ration (Rb/Ra)	Gain	Gain <sup>2</sup> (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

Note: Ra & Rb are in  $k\Omega$ 

#### **ANA IN Amplifier Gain Settings**

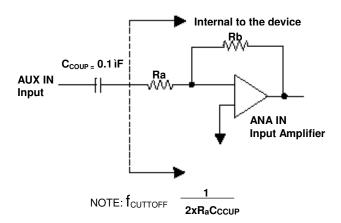
Setting <sup>(1)</sup>	0TLP Input	CFG0		Gain <sup>(2)</sup>	Array	Speaker
	$V_{P-P}^{(3)}$	AIG1	AIG0		In/Out V <sub>P-P</sub>	Out $V_{P-P}^{(4)}$
6 dB	1.110	0	0	0.625	0.694	2.22
9 dB	0.785	0	1	0.883	0.694	2.22
12 dB	0.555	1	0	1.250	0.694	2.22
15 dB	0.393	1	1	1.767	0.694	2.22

- 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)

### **AUX IN (Auxiliary Input)**

The AUX IN is an additional audio input to the ISD5100-Series, such as from the microphone circuit in a mobile phone "car kit." This input has a nominal 694 mV p-p level at its minimum gain setting (0 dB). See the AUX IN Amplifier Gain Settings table on page 56. Additional gain is available in 3 dB steps (controlled by the I<sup>2</sup>C interface) up to 9 dB.

**AUX IN Input Modes** 



Gain Setting	Resistor Ratio (Rb/Ra)	Gain	Gain <sup>(2)</sup> (dB)
00	40.1 / 40.1	1.0	0
01	47.0 / 33.2	1.414	3
10	53.5 / 26.7	2.0	6
11	59.2 / 21	2.82	9

Note: Ra & Rb are in  $k\Omega$ 

#### **AUX IN Amplifier Gain Settings**

Setting <sup>(1)</sup>	0TLP Input	CFG0		Gain <sup>(2)</sup>	Array	Speaker
	$V_{P-P}^{(3)}$	AXG1	AXG0	-	In/Out V <sub>P-P</sub>	Out V <sub>P-P</sub> <sup>(4)</sup>
0 dB	0.694	0	0	1.00	0.694	0.694
3 dB	0.491	0	1	1.41	0.694	0.694
6 dB	0.347	1	0	2.00	0.694	0.694
9 dB	0.245	1	1	2.82	0.694	0.694

1. Gain from AUX IN to ANA OUT

2. Gain from AUX IN to ARRAY IN

3. OTLP Input is the reference Transmission Level Point that is used for testing. This level is typically 3 dB below clipping

4. Differential

### 6.6.3 Power and Ground Pins

#### V<sub>CCA</sub>, V<sub>CCD</sub> (Voltage Inputs)

To minimize noise, the analog and digital circuits in the ISD5100 Series devices use separate power busses. These +3 V 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.

#### V<sub>SSA</sub>, V<sub>SSD</sub> (Ground Inputs)

The ISD5100 Series utilizes separate analog and digital ground busses. The analog ground (V<sub>SSA</sub>) 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<sub>SSD</sub>) 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<sub>SSA</sub> pins and the V<sub>SSD</sub> pin is less than 3 $\Omega$ . The backside of the die is connected to V<sub>SSD</sub> through the substrate resistance. In a chip-on-board design, the die attach area must be connected to V<sub>SSD</sub>.

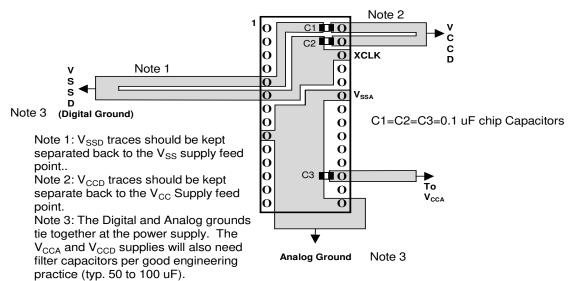
#### NC (Not Connect)

These pins should not be connected to the board at any time. Connection of these pins to any signal, ground or V<sub>CC</sub>, may result in incorrect device behavior or cause damage to the device.

# 6.6.4 PCB Layout Examples

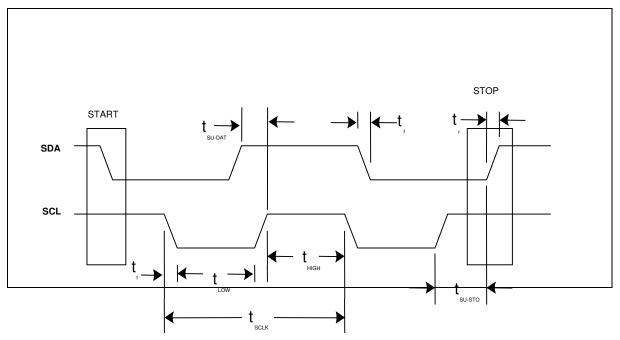
#### For SOIC package :

PC board traces and the three chip capacitors are on the bottom side of the board.



## 7. TIMING DIAGRAMS

7.1 I<sup>2</sup>C TIMING DIAGRAM



# I<sup>2</sup>C INTERFACE TIMING STANDARD-MODE FR SYMBOL MIN MAX MIN MAX

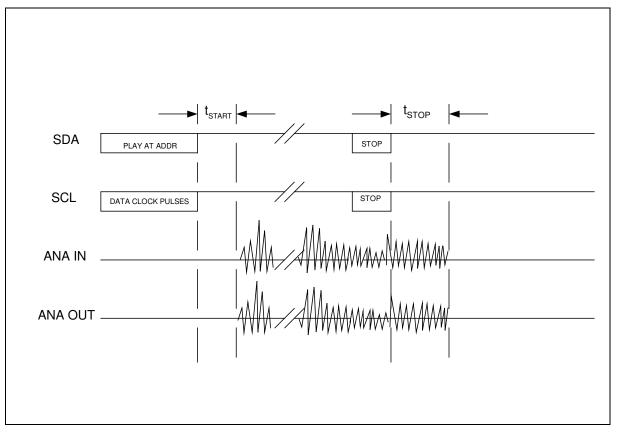
		MODE				
PARAMETER	SYMBOL	MIN.	MAX.	MIN.	MAX.	UNIT
SCL clock frequency	fsc∟	0	100	0	400	kHz
Hold time (repeated) START condition. After this period, the first clock pulse is generated	thd-sta	4.0	-	0.6	-	μs
LOW period of the SCL clock	t∟ow	4.7	-	1.3	-	μs
HIGH period of the SCL clock	tніgн	4.0	-	0.6	-	μs
Set-up time for a repeated START condition	tsu-sta	4.7	-	0.6	-	μs
Data set-up time	tsu-dat	250	-	100 <sup>(1)</sup>	-	ns
Rise time of both SDA and SCL signals	tr	-	1000	$20 + 0.1C_{b}^{(2)}$	300	ns
Fall time of both SDA and SCL signals	t <sub>f</sub>	-	300	$20 + 0.1C_{b}^{(2)}$	300	ns
Set-up time for STOP condition	tsu-sto	4.0	-	0.6	-	μs
Bus-free time between a STOP and START condition	<b>t</b> BUF	4.7	-	1.3	-	μs
Capacitive load for each bus line	Cb	-	400	-	400	pF
Noise margin at the LOW level for each connected device (including hysteresis)	VnL	0.1 V <sub>DD</sub>	-	0.1 V <sub>DD</sub>	-	V
Noise margin at the HIGH level for each connected device (including hysteresis)	VnH	0.2 V <sub>DD</sub>	-	0.2 V <sub>DD</sub>	-	V

 A Fast-mode l<sup>2</sup>C-interface device can be used in a Standard-mode l<sup>2</sup>C-interface system, but the requirement t<sub>SU;DAT</sub> ≥ 250 ns must then be met. This will automatically be the case if the device does not stretch the LOW period of the SCL signal.

If such a device does stretch the LOW period of the SCL signal, it must output the next data bit to the SDA line;  $t_{r max} + t_{SU;DAT} = 1000 + 250 = 1250$  ns (according to the Standard-mode  $l^2C$  -interface specification) before the SCL line is released.

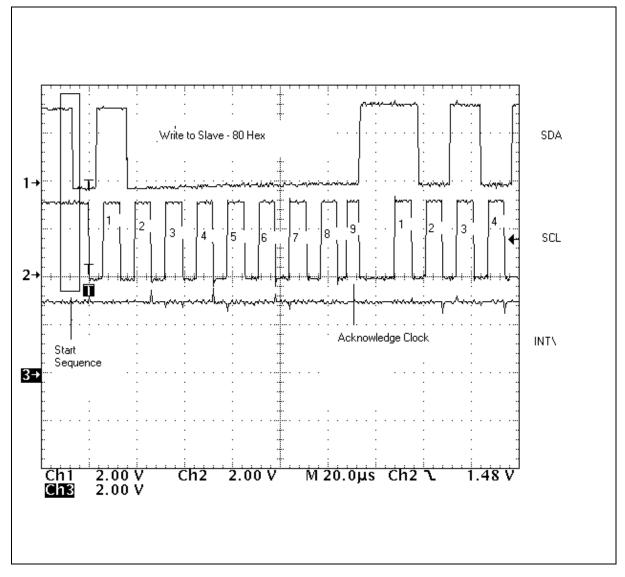
2. C<sub>b</sub> = total capacitance of one bus line in pF. If mixed with HS mode devices, faster fall-times are allowed.

#### 7.2 PLAYBACK AND STOP CYCLE



# ISD5100





### 8. ABSOLUTE MAXIMUM RATINGS

#### ABSOLUTE MAXIMUM RATINGS (PACKAGED PARTS)<sup>(1)</sup>

Condition	Value
Junction temperature	150ºC
Storage temperature range	-65ºC to +150ºC
Voltage Applied to any pins	(Vss - 0.3V) to (Vcc + 0.3V)
Voltage applied to any pin (Input current limited to +/-20 mA)	(Vss – 1.0V) to (Vcc + 1.0V)
Lead temperature (soldering - 10 seconds)	300°C
V <sub>CC</sub> - V <sub>SS</sub>	-0.3V to +5.5V

<sup>1.</sup> Stresses above those listed may cause permanent damage to the device. Exposure to the absolute maximum ratings may affect device reliability. Functional operation is not implied at these conditions.

### ABSOLUTE MAXIMUM RATINGS (DIE)<sup>(1)</sup>

Condition	Value
Junction temperature	150ºC
Storage temperature range	-65ºC to +150ºC
Voltage Applied to any pads	(V <sub>SS</sub> - 0.3V) to (V <sub>CC</sub> + 0.3V)
V <sub>CC</sub> - V <sub>SS</sub>	-0.3V to +5.5V

<sup>1.</sup> Stresses above those listed may cause permanent damage to the device. Exposure to the absolute maximum ratings may affect device reliability. Functional operation is not implied at these conditions.

# **OPERATING CONDITIONS (PACKAGED PARTS)**

Conditions	Values
Commercial operating temperature [1]	0°C to +70°C
Supply voltage (Vcc) [2]	+2.7V to +3.3V
Ground voltage (Vss) [3]	0V
Voltage Applied to any pins	(Vss - 0.3V) to (Vcc + 0.3V)

Cond	Values	
Industrial operating temperature	-40°C to +85°C	
Supply voltage (Vcc) [2]	Supply voltage (V <sub>CC</sub> ) <sup>[2]</sup> ISD5102, ISD5104, ISD5116	
	+2.7V to +3.6V	
Ground voltage (Vss) [3]	0V	
Voltage Applied to any pins		(Vss - 0.3V) to (Vcc + 0.3V)

# **OPERATING CONDITIONS (DIE)**

Conditions	Values
Die operating temperature range [1]	0°C to +50°C
Supply voltage (V <sub>CC</sub> ) [2]	+2.7V to +3.3V
Ground voltage (V <sub>SS</sub> ) <sup>[3]</sup>	0V
Voltage Applied to any pads	(V_{SS} - 0.3V) to (V_{CC} + 0.3V)

[1] Case temperature [2]  $V_{CC} = V_{CCA} = V_{CCD}$  [3]  $V_{SS} = V_{SSA} = V_{SSD}$ 

## 9. ELECTRICAL CHARACTERISTICS

#### 9.1. GENERAL PARAMETERS

Symbol	Parameters	Min <sup>(2)</sup>	<b>Typ</b> <sup>(1)</sup>	Max <sup>(2)</sup>	Units	Conditions
VIL	Input Low Voltage			$V_{CC} \ge 0.2$	V	
VIH	Input High Voltage	$V_{CC} \ge 0.8$			V	
V <sub>OL</sub>	SCL, SDA Output Low Voltage			0.4	V	$I_{OL} = 3 \ \mu A$
V <sub>IL2V</sub>	Input low voltage for 2V interface			0.4	V	Apply only to SCL, SDA
VIH2V	Input high voltage for 2V interface	1.6			V	Apply only to SCL, SDA
V <sub>OL1</sub>	RAC, INT Output Low Voltage			0.4	V	I <sub>OL</sub> = 1 mA
V <sub>OH</sub>	Output High Voltage	Vcc-0.4			V	I <sub>OL</sub> = -10 μA
lcc	V <sub>CC</sub> Current (Operating)					
	- Playback		15	25	mA	No Load <sup>(3)</sup>
	- Record		30	40	mA	No Load <sup>(3)</sup>
	- Feedthrough		12	15	mA	No Load <sup>(3)</sup>
I <sub>SB</sub>	Vcc Current (Standby)		1	10	μA	[3]
lı∟	Input Leakage Current			±1	μA	

<sup>[1]</sup> Typical values:  $T_A = 25^{\circ}C$  and Vcc = 3.0 V.

<sup>[2]</sup> All min/max limits are guaranteed by Nuvoton via electrical testing or characterization. Not all specifications are 100 percent tested.

<sup>[3]</sup> All V<sub>CC</sub> and V<sub>SS</sub> are connected appropriately and others are floated.

### 9.2. TIMING PARAMETERS

Symbol	Parameters	Min <sup>(2)</sup>		Ту	p <sup>(1)</sup>		Max <sup>(2)</sup>	Units	Conditions
Fs	Sampling Frequency			6 5	.0 .4 .3 .0			kHz kHz kHz kHz	(5) (5) (5) (5)
Fcf	Filter Knee 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)			2 2	.4 .7 .3 .7			kHz kHz kHz kHz	Knee Point <sup>(3)(7)</sup> Knee Point <sup>(3)(7)</sup> Knee Point <sup>(3)(7)</sup> Knee Point <sup>(3)(7)</sup>
Trec	Record Duration 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)		ISD5116 8.73 10.9 13.1 17.5	ISD5108 4.36 5.45 6.55 8.75	ISD5104 2.18 2.72 3.27 4.37	ISD5102 1.08 1.35 1.63 2.18		min min min min	(6) (6) (6) (6)
Tplay	Playback Duration 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)		ISD5116         ISD5108         ISD5104         ISD5102           8.73         4.36         2.18         1.08           10.9         5.45         2.72         1.35           13.1         6.55         3.27         1.63           17.5         8.75         4.37         2.18					min min min min	(6) (6) (6) (6)
Tpud	Power-Up Delay 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)				1 1 1			msec msec msec msec	
Tstop/ pause	Stop or Pause Record or Play 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)			4 4	2 0 8 4			msec msec msec msec	
T <sub>RAC</sub>	RAC Clock Period 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)			32 38	56 20 34 12			msec msec msec msec	(9) (9) (9) (9)
Tracl	RAC Clock Low Time 8.0 kHz (sample rate) 6.4 kHz (sample rate) 5.3 kHz (sample rate) 4.0 kHz (sample rate)			1 12	3 0 2.1 6			msec msec msec msec	
TRACM	RAC Clock Period in Message Cueing Mode 8.0 kHz (sample rate)			50	00			µsec	

	6.4 kHz (sample rate)	625		µsec	
	5.3 kHz (sample rate)	750		μsec	
	4.0 kHz (sample rate)	1000		μsec	
TRACML	RAC Clock Low Time in Message Cueing Mode				
	8.0 kHz (sample rate)	15.6		µsec	
	6.4 kHz (sample rate)	19.5		, µsec	
	5.3 kHz (sample rate)	23.4		μsec	
	4.0 kHz (sample rate)	31.2		μsec	
TRACE	RAC Clock Period in Digital Erase Mode				
	8.0 kHz (sample rate)	1.25		msec	
	6.4 kHz (sample rate)	1.56		msec	
	5.3 kHz (sample rate)	1.87		msec	
	4.0 kHz (sample rate)	2.50		msec	
TRACEL	RAC Clock Low Time in Digital Erase mode				
	8.0 kHz (sample rate)	0.25		msec	
	6.4 kHz (sample rate)	0.31		msec	
	5.3 kHz (sample rate)	0.37		msec	
	4.0 kHz (sample rate)	0.50		msec	
THD	Total Harmonic Distortion				@1 kHz at
	ANA IN to ARRAY,	1	2	%	0TLP, sample
	ARRAY to SPKR	1	2	%	rate = 5.3 kHz

#### 9.3. ANALOG PARAMETERS

# **MICROPHONE INPUT<sup>(14)</sup>**

Symbol	Parameters	Min (2)	<b>Typ</b> (1)(14)	Max (2)	Units	Conditions
V <sub>MIC+/-</sub>	MIC +/- Input Voltage			300	mV	Peak-to-Peak <sup>(4)(8)</sup>
VMIC (0TLP)	MIC +/- input reference transmission level point (0TLP)		208		mV	Peak-to-Peak <sup>(4)(10)</sup>
Аміс	Gain from MIC +/- input to ANA OUT	5.5	6.0	6.5	dB	1 kHz at $V_{MIC (0TLP)}^{(4)}$
AMIC (GT)	MIC +/- Gain Tracking		+/- 0.1		dB	1 kHz, +3 to –40 dB 0TLP Input
Rмic	Microphone input resistance		10		kΩ	MIC- and MIC+ pins
AAGC	Microphone AGC Amplifier Range	6		40	dB	Over 3-300 mV Range

## **ANA IN**<sup>(14)</sup>

Symbol	Parameters	<b>Min</b> (2)	<b>Typ</b> (1)(14)	Max (2)	Units	Conditions
Vana in	ANA IN Input Voltage			1.6	V	Peak-to-Peak (6 dB gain setting)
VANA IN (0TLP)	ANA IN (0TLP) Input Voltage		1.1		V	Peak-to-Peak (6 dB gain setting) <sup>(10)</sup>
AANA IN (sp)	Gain from ANA IN to SP+/-		+6 to +15		dB	4 Steps of 3 dB
Aana in (Aux Out)	Gain from ANA IN to AUX OUT		-4 to +5		dB	4 Steps of 3 dB
AANA IN (GA)	ANA IN Gain Accuracy	-0.5		+0.5	dB	(11)
AANA IN (GT)	ANA IN Gain Tracking		+/- 0.1		dB	1000 Hz, +3 to -45 dB 0TLP Input, 6 dB setting
Rana in	ANA IN Input Resistance (6 dB to +15 dB)		10 to 100		kΩ	Depending on ANA IN Gain

# AUX IN<sup>(14)</sup>

Symbol	Parameters	Min (2)	<b>Typ</b> (1)(14)	Max (2)	Unit s	Conditions
VAUX IN	AUX IN Input Voltage			1.0	V	Peak-to-Peak (0 dB gain setting)
VAUX IN (0TLP)	AUX IN (0TLP) Input Voltage		694.2		mV	Peak-to-Peak (0 dB gain setting)
Aaux in (Ana Out)	Gain from AUX IN to ANA OUT		0 to +9		dB	4 Steps of 3 dB
AAUX IN (GA)	AUX IN Gain Accuracy	-0.5		+0.5	dB	(11)
Aaux in (GT)	AUX IN Gain Tracking		+/- 0.1		dB	1000 Hz, +3 to –45 dB 0TLP Input, 0 dB setting
Raux in	AUX IN Input Resistance		10 to 100		kΩ	Depending on AUX IN Gain

# SPEAKER OUTPUTS<sup>(14)</sup>

Symbol	Parameters	<b>Min</b> (2)	<b>Typ</b> (1)(14)	Max (2)	Units	Conditions
VSPHG	SP+/- Output Voltage (High Gain Setting)			3.6	V	Peak-to-Peak, differential load = $150\Omega$ , OPA1, OPA0 = 01
Rsplg	SP+/- Output Load Imp. (Low Gain)	8			Ω	OPA1, OPA0 = 10
Rsphg	SP+/- Output Load Imp. (High Gain)	70	150		Ω	OPA1, OPA0 = 01
Csp	SP+/- Output Load Cap.			100	pF	
VSPAG	SP+/- Output Bias Voltage (Analog Ground)		1.2		VDC	
VSPDCO	Speaker Output DC Offset			+/- 100	mV DC	With ANA IN to Speaker, ANA IN AC coupled to $V_{\mbox{\scriptsize SSA}}$
ICNANA IN/(SP+/-)	ANA IN to SP+/- Idle Channel Noise			-65	dB	Speaker Load = $150\Omega^{(12)(13)}$
CrT(SP+/- )/ANA OUT	SP+/- to ANA OUT Cross Talk			-65	dB	1 kHz 0TLP input to ANA IN, with MIC+/- and AUX IN AC coupled to Vss, and measured at ANA OUT feed through mode $^{\rm (12)}$
PSRR	Power Supply Rejection Ratio		-55		dB	Measured with a 1 kHz, 100 mV p-p sine wave input at $V_{\rm CC}$ and $V_{\rm CC}$ pins
FR	Frequency Response (300-3400 Hz)		<u>+</u> 0.5		dB	With 0TLP input to ANA IN, 6 dB setting <sup>(12)</sup> Guaranteed by design
Poutlg	Power Output (Low Gain Setting)	23.5			mW RMS	Differential load at $8\Omega$

;	SINAD	SINAD ANA IN to SP+/-	62.5		dB	0TLP ANA In input minimum gain, 150 $\Omega$ load $^{(12)(13)}$

# ANA OUT <sup>(14)</sup>

Symbol	Parameters	<b>Min</b> (2)	<b>Typ</b> e (1)(14)	Max (2)	Units	Conditions
SINAD	SINAD, MIC IN to ANA OUT	62.5			dB	Load = $5k\Omega^{(12)(13)}$
SINAD	SINAD, AUX IN to ANA OUT (0 to 9 dB)	62.5			dB	Load = $5k\Omega^{(12)(13)}$
ICO <sub>NIC/ANA</sub> OUT	Idle Channel Noise – Microphone			-65	dB	Load = $5k\Omega^{(12)(13)}$
ICN AUX IN/ANA OUT	Idle Channel Noise – AUX IN (0 to 9 dB)			-65	dB	Load = $5k\Omega^{(12)(13)}$
PSRR (ANA OUT)	Power Supply Rejection Ratio		-40		dB	Measured with a 1 kHz, 100 mV $_{\text{P-P}}$ sine wave to $V_{\text{CCA}},V_{\text{CCD}}$ pins
VBIAS	ANA OUT+ and ANA OUT-		1.2		VDC	Inputs AC coupled to V <sub>SSA</sub>
VOFFSET	ANA OUT+ to ANA OUT-			+/- 100	mV DC	Inputs AC coupled to V <sub>SSA</sub>
RL	Minimum Load Impedance	5			kΩ	Differential Load
F <sub>R</sub>	Frequency Response (300- 3400 Hz)		<u>+</u> 0.5		dB	0TLP input to MIC+/- in feedthrough mode. 0TLP input to AUX IN in feedthrough mode <sup>(12)</sup>
C <sub>R</sub> T <sub>ANA</sub> OUT/(SP+/-)	ANA OUT to SP+/- Cross Talk			-65	dB	1 kHz 0TLP output from ANA OUT, with ANA IN AC coupled to $V_{SSA}$ , and measured at SP+/- <sup>(12)</sup>
CrTana out/aux out	ANA OUT to AUX OUT Cross Talk			-65	dB	1 kHz 0TLP output from ANA OUT, with ANA IN AC coupled to $V_{\text{SSA}}$ and measured at AUX $\text{OUT}^{(12)}$

# AUX OUT<sup>(14)</sup>

Symbol	Parameters	Min (2)	<b>Typ</b> (1(14))	Max (2)	Units	Conditions
Vaux out	AUX OUT – Maximum Output Swing			1.0	V	5kΩ Load
RL	Minimum Load Impedance	5			KΩ	
CL	Maximum Load Capacitance			100	pF	
VBIAS	AUX OUT		1.2		VDC	
SINAD	SINAD – ANA IN to AUX OUT	62.5			dB	0TLP ANA IN input, minimum gain, 5k load <sup>(12)(13)</sup>
ICN(AUX OUT)	Idle Channel Noise – ANA IN to AUX OUT			-65	dB	Load= $5k\Omega^{(12)(13)}$
CrTaux out/ana out	AUX OUT to ANA OUT Cross Talk			-65	dB	1 kHz 0TLP input to ANA IN, with MIC +/- and AUX IN AC coupled to $V_{SSA}$ , measured at SP+/-, load = 5k $\Omega$ . Referenced to nominal 0TLP @ output

# VOLUME CONTROL<sup>(14)</sup>

Symbol	Parameters	Min (2)	<b>Typ</b> (1)(14)	Max (2)	Unit s	Conditions
Аоит	Output Gain		-28 to 0		dB	8 steps of 4 dB, referenced to output
	Tolerance for each step	-1.0		+1.0	dB	ANA IN 1.0 kHz 0TLP, 6 dB gain setting measured differentially at SP+/-

#### Conditions

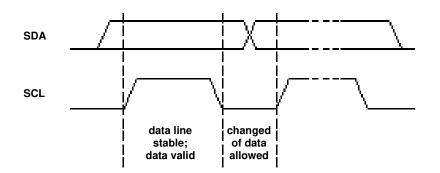
- 1. Typical values:  $T_A = 25^{\circ}C$  and Vcc = 3.0V.
- 2. All min/max limits are guaranteed by Nuvoton via electrical testing or characterization. Not all specifications are 100 percent tested.
- 3. Low-frequency cut off depends upon the value of external capacitors (see Pin Descriptions).
- 4. Differential input mode. Nominal differential input is 208 mV p-p. (0TLP)
- 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).
- 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).
- 7. Filter specification applies to the low pass filter.
- 8. For optimal signal quality, this maximum limit is recommended.
- 9. When a record command is sent,  $T_{RAC} = T_{RAC} + T_{RACL}$  on the first page addressed.
- 10. The maximum signal level at any input is defined as 3.17 dB 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 the <u>ANA IN table</u> and <u>AUX IN table</u> on pages 54 and 55 respectively.
- 12. 0TLP is the reference test level through inputs and outputs. See the <u>ANA IN table</u> and <u>AUX IN table</u> on pages 54 and 55 respectively.
- 13. Referenced to 0TLP input at 1 kHz, measured over 300 to 3,400 Hz bandwidth.
- 14. For die, only typical values are applicable.

### 9.4. CHARACTERISTICS OF THE I<sup>2</sup>C SERIAL INTERFACE

The I<sup>2</sup>C interface is for bi-directional, two-line communication between different ICs or modules. The two lines are a serial data line (SDA) and a serial clock line (SCL). Both lines must be connected to a positive supply via a pull-up resistor. Data transfer may be initiated only when the interface bus is not busy.

#### Bit transfer

One data bit is transferred during each clock pulse. The data on the SDA line must remain stable during the HIGH period of the clock pulse, as changes in the data line at this time will be interpreted as a control signal.

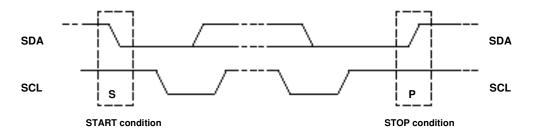


#### Start and stop co

#### Bit transfer on the I<sup>2</sup>C-

Both data and clo

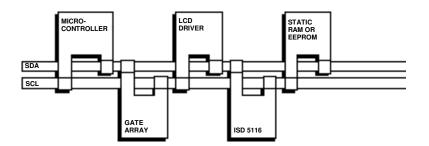
H-to-LOW transition of the data line while the clock is HIGH is defined as the start condition (S). A LOW-to-HIGH transition of the data line while the clock is HIGH is defined as the stop condition (P).



Definition of START and STOP conditions

### **System Configuration**

A device generating a message is a 'transmitter'; a device receiving a message is the 'receiver'. The device that controls the message I sthe 'master' and the devices that are controlled by the master are the 'slaves'.

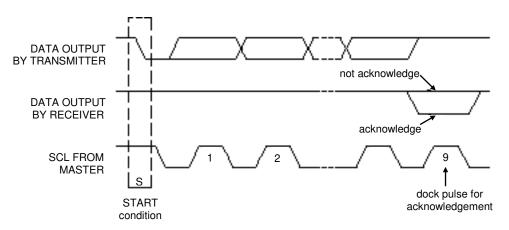


#### Acknowledge

Example of an I<sup>2</sup>C-bus configuration using two microcontrollers

The number of data bytes transferred between the start and stop conditions from transmitter to receiver is unlimited. Each byte of eight bits is followed by an acknowledge bit. The acknowledge bit is a HIGH level signal put on the interface bus by the transmitter during which time the master generates an extra acknowledge related clock pulse. A slave receiver which is addressed must generate an acknowledge after the reception of each byte. In addition, a master receiver must generate an acknowledge after the reception of each byte that has been clocked out of the slave transmitter.

The device that acknowledges must pull down the SDA line during the acknowledge clock pulse so that the SDA line is stable LOW during the HIGH period of the acknowledge related clock pulse (set-up and hold times must be taken into consideration). A master receiver must signal an end of data to the transmitter by not generating an acknowledge on the last byte that has been clocked out of the slave. In this event, the transmitter must leave the data line HIGH to enable the master to generate a stop condition.



Acknowledge on the I<sup>2</sup>C-bus

#### 9.5. I<sup>2</sup>C PROTOCOL

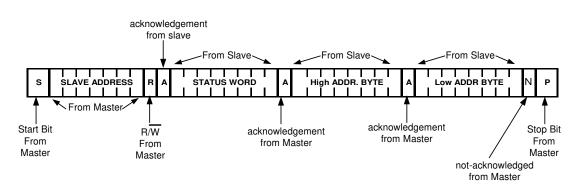
Since the I2C protocol allows multiple devices on the bus, each device must have an address. This address is known as a "Slave Address". A Slave Address consists of 7 bits, followed by a single bit that indicates the direction of data flow. This single bit is 1 for a Write cycle, which indicates the data is being sent from the current bus master to the device being addressed. This single bit is a 0 for a Read cycle, which indicates that the data is being sent from the device being addressed to the current bus master. For example, the valid Slave Addresses for the ISD5100 Series device, for both Write and Read cycles, are shown in section 7.3.1 on page 13 of this datasheet.

Before any data is transmitted on the I2C interface, the current bus master must address the slave it wishes to transfer data to or from. The Slave Address is always sent out as the 1<sup>st</sup> byte following the Start Condition sequence. An example of a Master transmitting an address to a ISD5100 Series slave is shown below. In this case, the Master is writing data to the slave and the R/W bit is "0", i.e. a Write cycle. All the bits transferred are from the Master to the Slave, except for the indicated Acknowledge bits. The following example details the transfer explained in <u>section 7.3.1-2-3</u> on pages 13-20 of this datasheet.

#### acknowledgement acknowledgement acknowledgement acknowledgement from slave from slave from slave from slave COMMAND BYTE High ADDR. BYTE Low ADDR. BYTE SLAVE ADDRESS w Р Start Bit Stop Bit R/W

A common procedure in the ISD5100 Series is the reading of the Status Bytes. The Read Status condition in the ISD5100 Series is triggered when the Master addresses the chip with its proper Slave Address, immediately followed by the R/W bit set to a "1" and without the Command Byte being sent. This is an example of the Master sending to the Slave, immediately followed by the Slave sending data back to the Master. The "N" not-acknowledge cycle from the Master ends the transfer of data from the Slave. The following example details the transfer explained in <u>section 7.3.1</u> on page 13 of this datasheet.

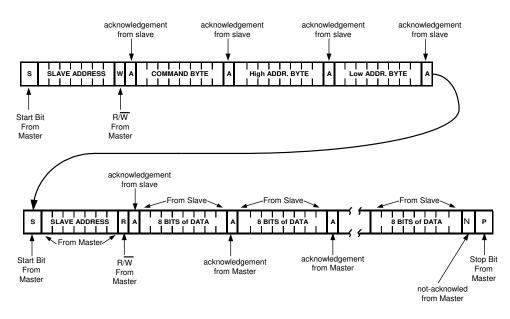
#### Master Transmits to Slave Receiver (Write) Mode



Master Reads from Slave immediately after first byte (Read Mode)

Another common operation in the ISD5100 Series is the reading of digital data from the chip's memory array at a specific address. This requires the I<sup>2</sup>C interface Master to first send an address to the ISD5100 Series Slave device, and then receive data from the Slave in a single I<sup>2</sup>C operation. To accomplish this, the data direction R/W bit must be changed in the middle of the command. The following example shows the Master sending the Slave address, then sending a Command Byte and 2 bytes of address data to the ISD5100-Series, and then immediately changing the data direction and reading some number of bytes from the chip's digital array. An unlimited number of bytes can be read in this operation. The "N" not-acknowledge cycle from the Master forces the end of the data transfer from the Slave. The following example details the transfer explained in <u>section 7.5.4</u> on page 47 of this datasheet.

# Master Reads from the Slave after setting data address in Slave (Write data address, READ Data)



### **10. TYPICAL APPLICATION CIRCUIT**

The following typical application example on ISD5100 series is for references only. They make no representation or warranty that such applications shall be suitable for the use specified. It's customer's obligation to verify the design in its own system for the functionalities, voice quality, current consumption, and etc.

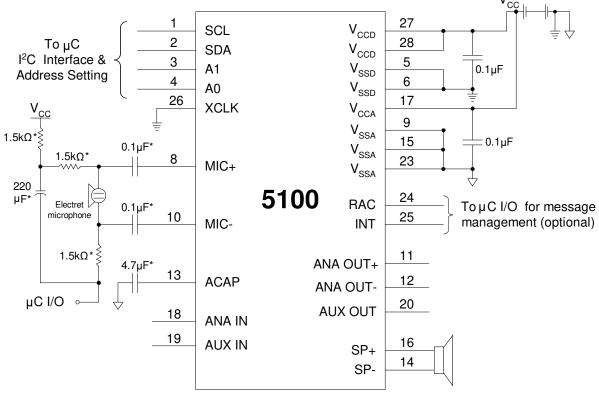
In addition, the below notes apply to the following application examples:

The suggested values are for references only. Depending on system requirements, they can be adjusted for functionalities and better performance.

It is important to have a separate path for each ground and power back to the related terminals to minimize the noise. Besides, the power supplies should be decoupled as close to the device as possible.

Also, it is crucial to follow good audio design practices in layout and power supply decoupling. See recommendations in Application Notes from our websites.

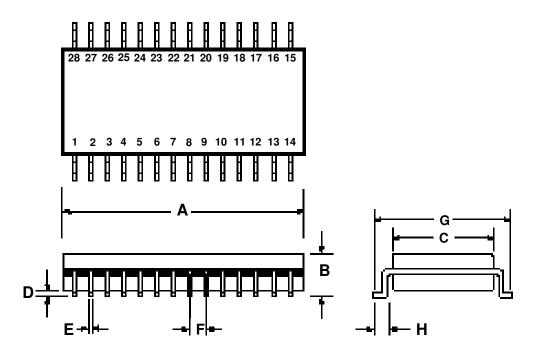
Example #1: Recording via microphone



SOIC / PDIP

## **11. PACKAGE SPECIFICATION**

11.1. 28-LEAD 300-MIL PLASTIC SMALL OUTLINE INTEGRATED CIRCUIT (SOIC)

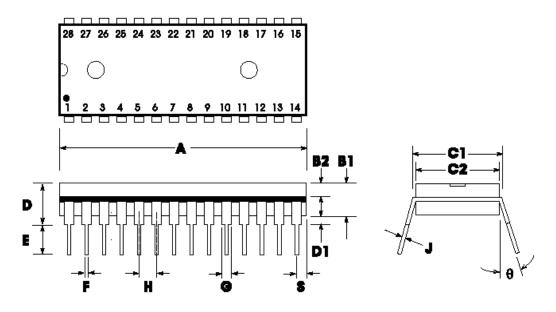


Plastic Small Outline Integrated Circuit (SOIC) Dimensions

	INCHES			MILLIMETERS		
	Min	Nom	Max	Min	Nom	Max
Α	0.701	0.706	0.711	17.81	17.93	18.06
В	0.097	0.101	0.104	2.46	2.56	2.64
С	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
Н	0.024	0.032	0.040	0.61	0.81	1.02

*Note:* Lead coplanarity to be within 0.004 inches.

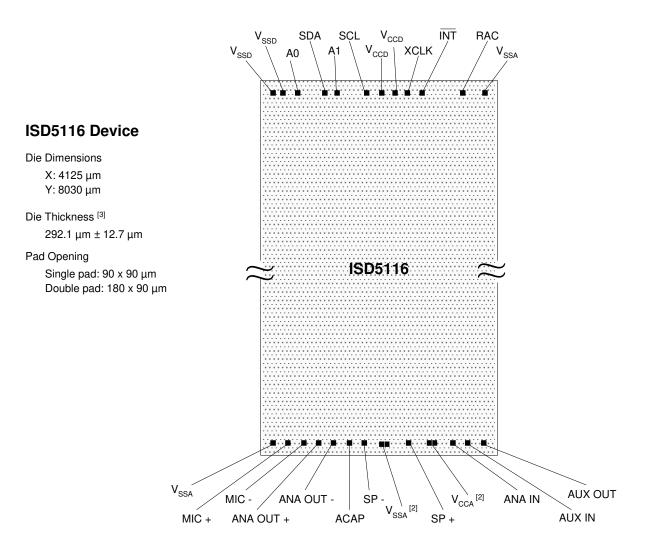
### 11.2. 28-LEAD 600-MIL PLASTIC DUAL INLINE PACKAGE (PDIP)



	INCHES			MILLIMETERS		
	Min	Nom	Max	Min	Nom	Max
А	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	1.65
Н		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
0	0°		15°	0°		15°



### 11.3. ISD5116 DIE INFORMATION



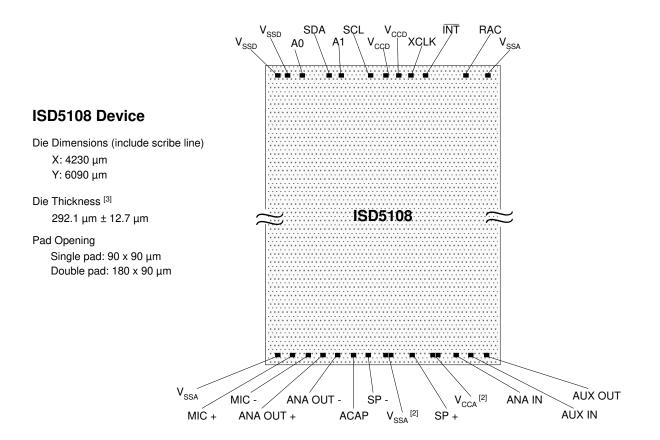
- 1. The backside of die is internally connected to Vss. It MUST NOT be connected to any other potential or damage may occur.
- 2. Double bond recommended, if treated as single doubled-pad.
- 3. This figure reflects the current die thickness. Please contact Nuvoton as this thickness may change in the future.

## ISD5116 Pad Coordinates

Pad	Pad Name	X Axis	Y Axis
V <sub>SSA</sub>	Analog Ground	1879.45	3848.65
RAC	Row Address Clock	1536.20	3848.65
INT	Interrupt	787.40	3848.65
XCLK	External Clock Input	475.60	3848.65
V <sub>CCD</sub>	Digital Supply Voltage	288.60	3848.65
V <sub>CCD</sub>	Digital Supply Voltage	73.20	3848.65
SCL	Serial Clock Line	-201.40	3848.65
A1	Address 1	-560.90	3848.65
SDA	Serial Data Address	-818.20	3848.65
A0	Address 0	-1369.40	3848.65
V <sub>SSD</sub>	Digital Ground	-1671.30	3848.65
V <sub>SSD</sub>	Digital Ground	-1842.90	3848.65
V <sub>SSA</sub>	Analog Ground	-1948.00	-3841.60
MIC+	Non-inverting Microphone Input	-1742.20	-3841.60
MIC-	Inverting Microphone Input	-1509.70	-3841.60
ANA OUT+	Non-inverting Analog Output	-1248.00	-3841.60
ANA OUT-	Inverting Analog Output	-913.80	-3841.60
ACAP	AGC/AutoMute Cap	-626.50	-3841.60
SP-	Speaker Negative	-130.70	-3841.60
V <sub>SSA</sub>	Analog Ground	202.90	-3841.60
V <sub>SSA</sub>	Analog Ground	292.90	-3841.60
SP+	Speaker Positive	626.50	-3841.60
VCCA	Analog Supply Voltage	960.10	-3841.60
V <sub>CCA</sub>	Analog Supply Voltage	1050.10	-3841.60
ANA IN	Analog Input	1257.40	-3841.60
AUX IN	Auxiliary Input	1523.00	-3841.60
AUX OUT	Auxiliary Output	1767.20	-3841.60

ISD5100

### 11.4. ISD5108 DIE INFORMATION



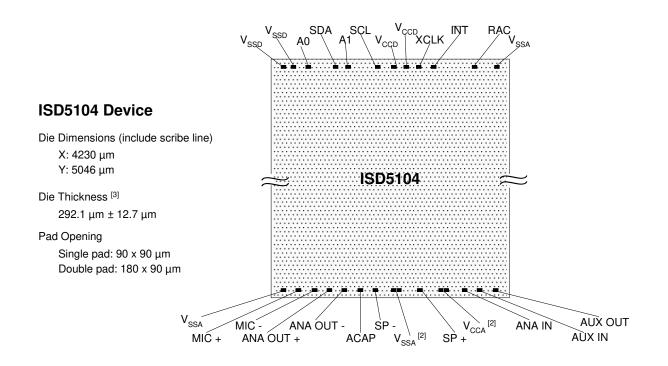
- 1. The backside of die is internally connected to Vss. It MUST NOT be connected to any other potential or damage may occur.
- 2. Double bond recommended, if treated as single doubled-pad.
- 3. This figure reflects the current die thickness. Please contact Nuvoton as this thickness may change in the future.

## ISD5108 Pad Coordinates

Pad	Pad Name	X Axis	Y Axis
Vssa	Analog Ground	1882.40	2820.65
RAC	Row Address Clock	1539.15	2820.65
INT	Interrupt	790.35	2820.65
XCLK	External Clock Input	478.55	2820.65
VCCD	Digital Supply Voltage	291.55	2820.65
VCCD	Digital Supply Voltage	76.15	2820.65
SCL	Serial Clock Line	-198.45	2820.65
A1	Address 1	-557.95	2820.65
SDA	Serial Data Address	-815.25	2820.65
A0	Address 0	-1366.45	2820.65
V <sub>SSD</sub>	Digital Ground	-1668.35	2820.65
V <sub>SSD</sub>	Digital Ground	-1839.95	2820.65
V <sub>SSA</sub>	Analog Ground	-1945.05	-2821.60
MIC+	Non-inverting Microphone Input	-1739.25	-2821.60
MIC-	Inverting Microphone Input	-1506.75	-2821.60
ANA OUT+	Non-inverting Analog Output	-1245.05	-2821.60
ANA OUT-	Inverting Analog Output	-910.85	-2821.60
ACAP	AGC/AutoMute Cap	-623.55	-2821.60
SP-	Speaker Negative	-127.75	-2821.60
Vssa	Analog Ground	205.85	-2821.60
Vssa	Analog Ground	295.85	-2821.60
SP+	Speaker Positive	629.45	-2821.60
Vcca	Analog Supply Voltage	963.05	-2821.60
V <sub>CCA</sub>	Analog Supply Voltage	1053.05	-2821.60
ANA IN	ANA IN Analog Input		-2821.60
AUX IN	Auxiliary Input	1525.95	-2821.60
AUX OUT	AUX OUT Auxiliary Output		-2821.60

ISD5100

#### 11.5. ISD5104 DIE INFORMATION



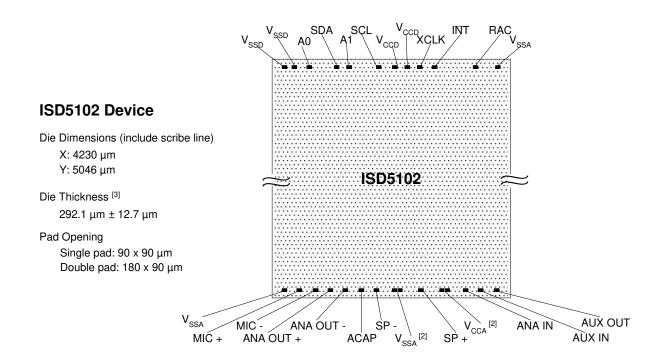
- 1. The backside of die is internally connected to Vss. It MUST NOT be connected to any other potential or damage may occur.
- 2. Double bond recommended, if treated as single doubled-pad.
- 3. This figure reflects the current die thickness. Please contact Nuvoton as this thickness may change in the future.

## ISD5104 Pad Coordinates

Pad	Pad Name	X Axis	Y Axis
V <sub>SSA</sub>	Analog Ground	1882.4	2306.65
RAC	Row Address Clock	1539.15	2306.65
INT	Interrupt	790.35	2306.65
XCLK	External Clock Input	478.55	2306.65
V <sub>CCD</sub>	Digital Supply Voltage	291.55	2306.65
V <sub>CCD</sub>	Digital Supply Voltage	76.15	2306.65
SCL	Serial Clock Line	-198.45	2306.65
A1	Address 1	-557.95	2306.65
SDA	Serial Data Address	-815.25	2306.65
A0	Address 0	-1366.45	2306.65
V <sub>SSD</sub>	Digital Ground	-1839.95	2306.65
V <sub>SSD</sub>	Digital Ground	-1668.35	2306.65
V <sub>SSA</sub>	Analog Ground	-1945.05	-2311.60
MIC+	Non-inverting Microphone Input	-1739.25	-2311.60
MIC-	Inverting Microphone Input	-1506.75	-2311.60
ANA OUT+	Non-inverting Analog Output	-1245.05	-2311.60
ANA OUT-	Inverting Analog Output	-910.85	-2311.60
ACAP	AGC/AutoMute Cap	-623.55	-2311.60
SP-	Speaker Negative	-127.75	-2311.60
V <sub>SSA</sub>	Analog Ground	205.85	-2311.60
V <sub>SSA</sub>	Analog Ground	295.85	-2311.60
SP+	Speaker Positive	629.45	-2311.60
V <sub>CCA</sub>	Analog Supply Voltage	963.05	-2311.60
VCCA	Analog Supply Voltage	1053.05	-2311.60
ANA IN	Analog Input	1260.35	-2311.60
AUX IN	Auxiliary Input	1525.95	-2311.60
AUX OUT	Auxiliary Output	1770.15	-2311.60

ISD5100

#### 11.6. ISD5102 DIE INFORMATION

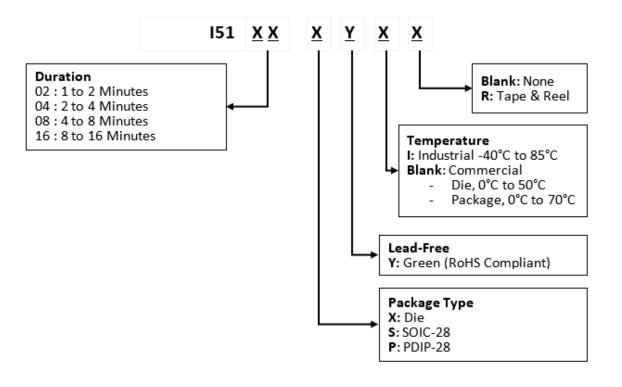


- 1. The backside of die is internally connected to Vss. It MUST NOT be connected to any other potential or damage may occur.
- 2. Double bond recommended, if treated as single doubled-pad.
- 3. This figure reflects the current die thickness. Please contact Nuvoton as this thickness may change in the future.

## ISD5102 Pad Coordinates

Pad	Pad Name	X Axis	Y Axis
V <sub>SSA</sub>	Analog Ground	1882.4	2306.65
RAC	Row Address Clock	1539.15	2306.65
INT	Interrupt	790.35	2306.65
XCLK	External Clock Input	478.55	2306.65
V <sub>CCD</sub>	Digital Supply Voltage	291.55	2306.65
V <sub>CCD</sub>	Digital Supply Voltage	76.15	2306.65
SCL	Serial Clock Line	-198.45	2306.65
A1	Address 1	-557.95	2306.65
SDA	Serial Data Address	-815.25	2306.65
A0	Address 0	-1366.45	2306.65
V <sub>SSD</sub>	Digital Ground	-1839.95	2306.65
V <sub>SSD</sub>	Digital Ground	-1668.35	2306.65
V <sub>SSA</sub>	Analog Ground	-1945.05	-2311.60
MIC+	Non-inverting Microphone Input	-1739.25	-2311.60
MIC-	Inverting Microphone Input	-1506.75	-2311.60
ANA OUT+	Non-inverting Analog Output	-1245.05	-2311.60
ANA OUT-	Inverting Analog Output	-910.85	-2311.60
ACAP	AGC/AutoMute Cap	-623.55	-2311.60
SP-	Speaker Negative	-127.75	-2311.60
V <sub>SSA</sub>	Analog Ground	205.85	-2311.60
V <sub>SSA</sub>	Analog Ground	295.85	-2311.60
SP+	Speaker Positive	629.45	-2311.60
V <sub>CCA</sub>	Analog Supply Voltage	963.05	-2311.60
VCCA	Analog Supply Voltage	1053.05	-2311.60
ANA IN	Analog Input	1260.35	-2311.60
AUX IN	Auxiliary Input	1525.95	-2311.60
AUX OUT	Auxiliary Output	1770.15	-2311.60

### **12.ORDERING INFORMATION**



Part Number	Duration	Package	Temperature	Notes
I5102X	1 to 2 Minutes	Die	0°C to 50°C	
I5102SY	1 to 2 Minutes	SOIC-28	0°C to 70°C	
I5102SYI	1 to 2 Minutes	SOIC-28	-40°C to 85°C	
I5104X	2 to 4 Minutes	Die	0°C to 50°C	
I5104SY	2 to 4 Minutes	SOIC-28	0°C to 70°C	
I5104SYR	2 to 4 Minutes	SOIC-28, Tape & Reel	0°C to 70°C	
I5108X	4 to 8 Minutes	Die	0°C to 50°C	
I5108SY	4 to 8 Minutes	SOIC-28	0°C to 70°C	
I5108SYI	4 to 8 Minutes	SOIC-28	-40°C to 85°C	
I5116X	8 to 16 Minutes	Die	0°C to 50°C	
I5116PY	8 to 16 Minutes	PDIP-28	0°C to 70°C	
I5116SY	8 to 16 Minutes	SOIC-28	0°C to 70°C	
I5116SYI	8 to 16 Minutes	SOIC-28	-40°C to 85°C	
I5116SYIR	8 to 16 Minutes	SOIC-28, Tape & Reel	-40°C to 85°C	

## **13. REVISION HISTORY**

VERSION	DATE	DESCRIPTION
1.0	Jul, 2004	Vcc: Industrial temp: $2.7 - 3.3V$ (5108: $2.7 - 3.6V$ ) Section 7.3.10 CFG0: Select $8\Omega$ speaker output Section 7.6.2 AuxOut: Correct parameter name to OPA1 Application diagram: Revise pin # on SDA & A1
1.1	Nov, 2004	Revise Operating conditions sections Add Pb-free option to Ordering Info
1.2	Apr, 2005	Revise Ordering Info section Update disclaim section
1.3	Oct, 2005	Revise Packaging information.
1.4	May, 2007	Remove leaded package offer Update XCLK description Revise application diagram Change logo
1.5	May 21, 2020	Update Document Format Remote TSOP Package
1.6	Jun 28, 2021	Update Ordering Information



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Insecure usage includes, but is not limited to: equipment for surgical implementation, atomic energy control instruments, airplane or spaceship instruments, the control or operation of dynamic, brake or safety systems designed for vehicular use, traffic signal instruments, all types of safety devices, and other applications intended to support or sustain life.

All Insecure Usage shall be made at customer's risk, and in the event that third parties lay claims to Nuvoton as a result of customer's Insecure Usage, customer shall indemnify the damages and liabilities thus incurred by Nuvoton.

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