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# **ISD5216**

**8 to 16 minutes**

**voice record/playback device**

**with integrated codec**



## 1. GENERAL DESCRIPTION

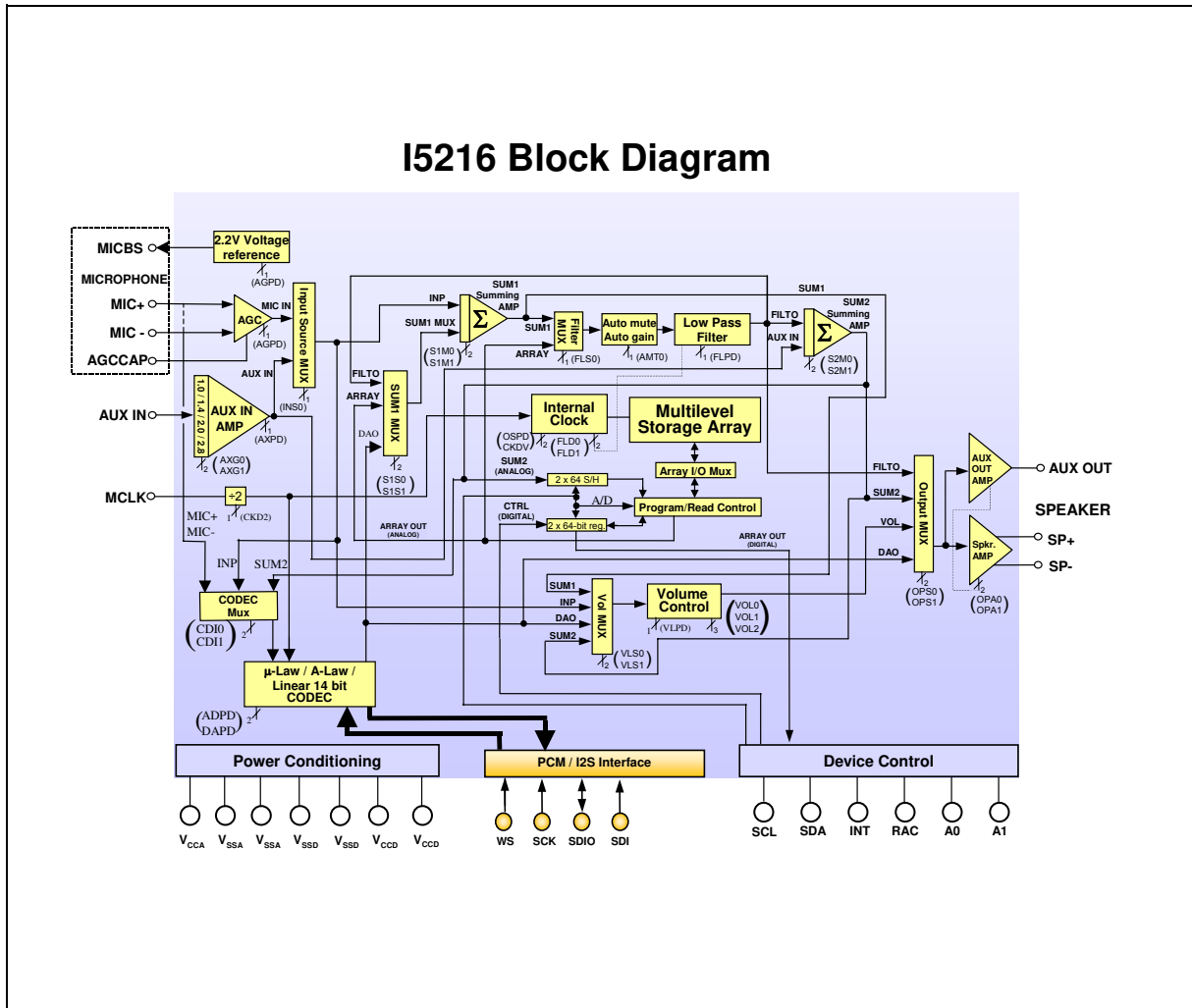
The ChipCorder ISD5216 is an 8 to 16 minute Voice and Data Record and Playback system with integrated Voice band CODEC. The device works on a single 2.7V to 3.3V supply, and has fully integrated system functions, including: AGC, microphone preamplifier, speaker driver, memory and CODEC. The CODEC meets the PCM conformance specification of the G.714 recommendation. Its  $\mu$ -Law and A-Law compander meets the specification of the ITU-T G.711 recommendation.

## 2. FEATURES

- Single Supply 2.7 to 3.3 Volt operation
- Voice and digital data record and playback system on a single chip
- Industry-leading sound quality
- Low voltage operation
- Message management
- Fully integrated system functions
- Flexible architecture
- Nonvolatile message storage
- Configurable ChipCorder sampling rates of 4 kHz, 5.3kHz, 6.4 kHz and 8kHz
- 8, 10, 12 and 16 minutes duration
- External or internal Voice recorder clock
- I<sup>2</sup>C serial interface (400kHz)
- Configurable analog paths
- 2.2V Microphone Bias Pin
- 100 year message retention (typical)
- 100K analog record cycles (typical)
- 10K digital record cycles (typical)
- Full-duplex (not in I<sup>2</sup>S mode) single channel speech CODEC with:
  - External 13.824 MHz, 27.648 MHz, 20.48 MHz or 40.96 MHz master clock
  - I<sup>2</sup>S and PCM digital audio interface ports
  - Serial transfer data rate from 64 to 3072 Kbps
  - Short and Long frame sync formats
  - 2s complement and signed magnitude data format
  - Complete  $\mu$ -Law and A-Law companding
  - Linear 14 bit  $\Delta\Sigma$  PCM CODEC-filter for A/D and D/A converter
  - 8 kHz or 44.1 kHz – 48 kHz digital audio sampling rate options
  - Analog receive and transmit gain adjust
  - Configurable setup through the I<sup>2</sup>C interface



3. BLOCK DIAGRAM





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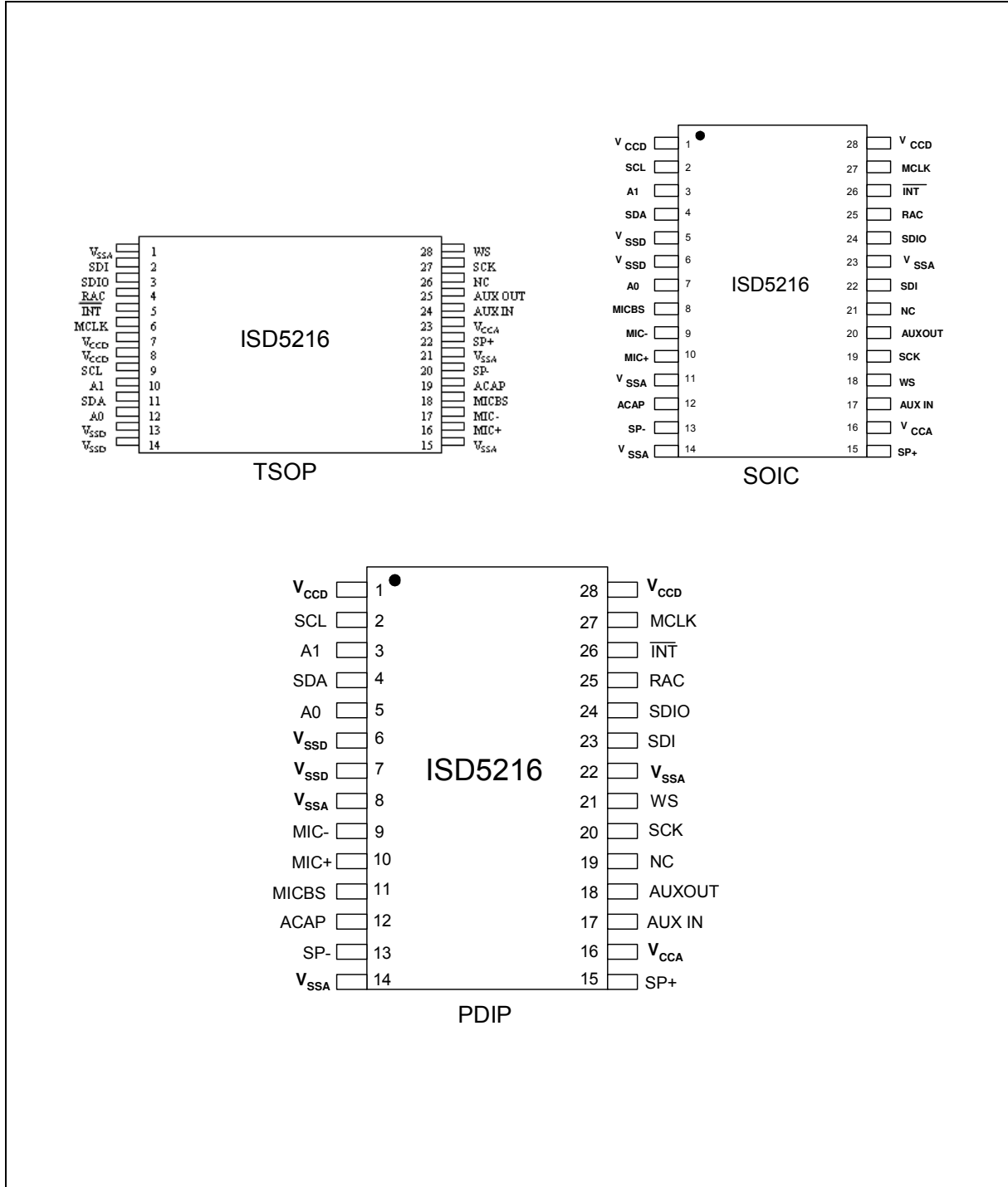
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5. PIN CONFIGURATION



Please note that the pin assignments are different for the PDIP and the SOIC packages.





## 6. PIN DESCRIPTION

Pin Name	Pin No. 28-pin TSOP	Pin No. 28-pin PDIP	Pin No. 28-pin SOIC	Functionality
RAC	4	25	25	Row Address Clock; an open drain output. The RAC pin goes LOW $T_{RACLO}$ <sup>1</sup> before the end of each row of memory, and returns HIGH at exactly the end of each row of memory.
$\overline{\text{INT}}$	5	26	26	Interrupt Output; an open drain output indicating 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.
MCLK	6	27	27	This pin allows the internal clock of the Voice record/playback system to be externally driven for enhanced timing precision. This pin is grounded for most applications. It is required for the CODEC operation.
SCL	9	2	2	Serial Clock Line is part of the I <sup>2</sup> C serial bus. It is used to clock the data into and out of the I <sup>2</sup> C interface.
SDA	11	4	4	Serial Data Line is part of the I <sup>2</sup> C serial bus. Data is passed between devices on the bus over this line.
A0	12	5	7	Input pin that supplies the LSB for the I <sup>2</sup> C Slave Address.
A1	10	3	3	Input pin that supplies the LSB +1 bit for the I <sup>2</sup> C Slave Address.
MIC+	16	9	10	Differential positive Input to the microphone amplifier.
MIC-	17	10	9	Differential negative Input to the microphone amplifier.
MICBS	18	11	8	Microphone Bias Voltage
ACAP	19	12	12	AGC Capacitor connection. Required for the on-chip AGC amplifier.
SP+	22	15	15	Differential Positive Speaker Driver Output.
SP-	20	13	13	Differential Negative Speaker Driver Output. When the speaker outputs are in use, the AUX OUT output is disabled.
AUX IN	24	17	17	Auxiliary Input.
AUX OUT	25	18	20	Auxiliary Output. This is one the analog outputs for the device. When this output is in use, the SP+ and SP- outputs are disabled.
SDI	2	23	22	Serial Digital Audio PCM Input.
SDIO	3	24	24	Serial Digital Audio PCM Output or I <sup>2</sup> S Input/Output.
WS	28	21	18	Digital audio PCM Frame sync (FS) or I <sup>2</sup> S Word Sync (WS).
SCK	27	20	19	Digital audio PCM or I <sup>2</sup> S Serial Clock.
V <sub>CCD</sub>	7,8	1,28	1,28	Positive Digital Supply pins. These pins carry noise generated by internal clocks in the chip. They must be carefully bypassed to Digital Ground to ensure correct device operation.
V <sub>SSD</sub>	13,14	6,7	5,6	Digital Ground pins.
V <sub>SSA</sub>	1,15,21	8,14,22	11,14,23	Analog Ground pins.
V <sub>CCA</sub>	23	16	16	Positive Analog Supply pin. This pin supplies the low level audio sections for the device. It should be carefully bypassed to Analog Ground to ensure correct device operation.
NC	26	19	21	No Connection

<sup>1</sup> See parameters section of the datasheet.



## 7. FUNCTIONAL DESCRIPTION

The ISD5216 ChipCorder Product provides high quality, fully integrated, single-chip Record/Playback solutions for 8- to 16-minute messaging applications that are ideal for use in PBX systems, cellular phones, automotive communications, GPS/navigation systems, and other portable products. The ISD5216 product is an enhancement to the ISD5116 architecture, providing: 1) A full-duplex Voice CODEC with  $\mu$ -Law and A-Law compander using the I<sup>2</sup>S and PCM interface ports; 2) A 2.2V microphone bias supply for reduced noise coupling. This supply can also be used to power down the external microphone with the system.

Analog functions and audio gating have also been integrated into the ISD5216 product to allow for easy interfacing with integrated chip sets on the market. Audio paths have been designed to enable full duplex conversation record, voice memo and answering machine (including outgoing message playback).

Logic Interface Options of 2.0V and 3.0V are supported by the ISD5216 to accommodate both portable communication (2.0- and 3.0-volt required) and automotive product customers (5.0-volt required).

Like other ChipCorder products, the ISD5216 integrates the sampling clock, anti-aliasing and smoothing filters, and multi-level storage array on a single chip. For enhanced voice features, the ISD5216 eliminates external circuitry by integrating automatic gain control (AGC), a power amplifier/speaker driver, volume control, summing amplifiers, analog switches, and a Voice CODEC. Input level adjustable amplifiers are also included, providing a flexible interface for multiple applications.

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

### *SPEECH/SOUND QUALITY*

The ISD5216 ChipCorder product can be software configured to operate at 4.0, 5.3, 6.4, and 8.0 kHz sampling frequencies, allowing the user a choice of speech quality options. Increasing the duration decreases the sampling frequency and bandwidth, which affects sound quality. The "Input Sample Duration" table below compares filter pass band and product durations.

### *DURATION*

To meet end-system requirements, the ISD5216 device is a single-chip solution, which provides 8 to 16 minutes of voice record and playback, depending on the sample rates defined by the customer's software.

Input Sample Rate to Duration Input Sample

Rate (kHz)	Duration <sup>1</sup> (Minutes)	Typical Filter Pass Band (kHz)
8.0	8 min 3 sec	3.7
6.4	10 min 4 sec	2.9
5.3	12 min 9 sec	2.5
4.0	16 min 6 sec	1.8



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

### ***FLASH STORAGE***

One of the benefits of Winbond's ChipCorder technology is the use of on-chip nonvolatile memory, which provides zero-power message storage. A 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 digital messages and over 100,000 times (typically) for analog messages.

Memory space can be allocated to either digital or analog storage, when recording. The system micro controller stores this information in the Message Address Table.

### ***MICROCONTROLLER INTERFACE***

The **ISD5216** is 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 to 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 pins can be connected to the microcontroller for enhanced interface: the RAC timing pin and the INT pin for interrupts to the controller. Communications with all of the internal registers is through the serial bus, as well as digital memory Read and Write operations.

### ***PROGRAMMING***

The ISD5216 series is also ideal for playback-only applications, whereas single or multiple messages may be played back when desired. Playback is controlled through the I<sup>2</sup>C port. Once the desired message configuration is created, duplicates can easily be generated via a Winbond or third-party programmer. For more information on available application tools and programmers, please see the Winbond web site at <http://www.winbond-usa.com/>.

### ***AUDIO PATHS***

The ISD5216 has extremely powerful audio routing functionality where all audio signals can be routed and multiplexed to multiple destinations. A few examples are

- Simultaneous recording of microphone input and CODEC DAC output for recording both parties of a phone call.



## 7.1. MEMORY ORGANIZATION

The ISD5216 memory array is arranged as 1888 rows (or pages) of 2048 bits, for a total memory of 3,866,624 bits. The primary addressing for the 2048 pages is handled by 11 bits of address data in the analog mode. At the 8 kHz sample rate, each page contains 256 milliseconds of audio. Thus, at 8 kHz there is actually room for 8 minutes and 3 seconds of audio.

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 (op code) at the time the data is written. A record of what is analog and what is digital, and where, is stored by the system microcontroller in the message address table (MAT). The MAT is a table kept in the microcontroller memory that defines the status of each message "block." It can be stored back into the ISD5216 if the power fails or the system is turned off. Use of this table allows for efficient message management. Segments of messages can be stored wherever there is available space in the memory array.

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, 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 issued, but continues until the 32-millisecond block is filled. Then a bit is placed into 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 acknowledgement. If data is coming in faster than it can be written, then the chip will not issue an acknowledgement to the host microcontroller until it is ready.

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<sup>2</sup>C port. (See [Digital Mode](#) on page 26 for details).

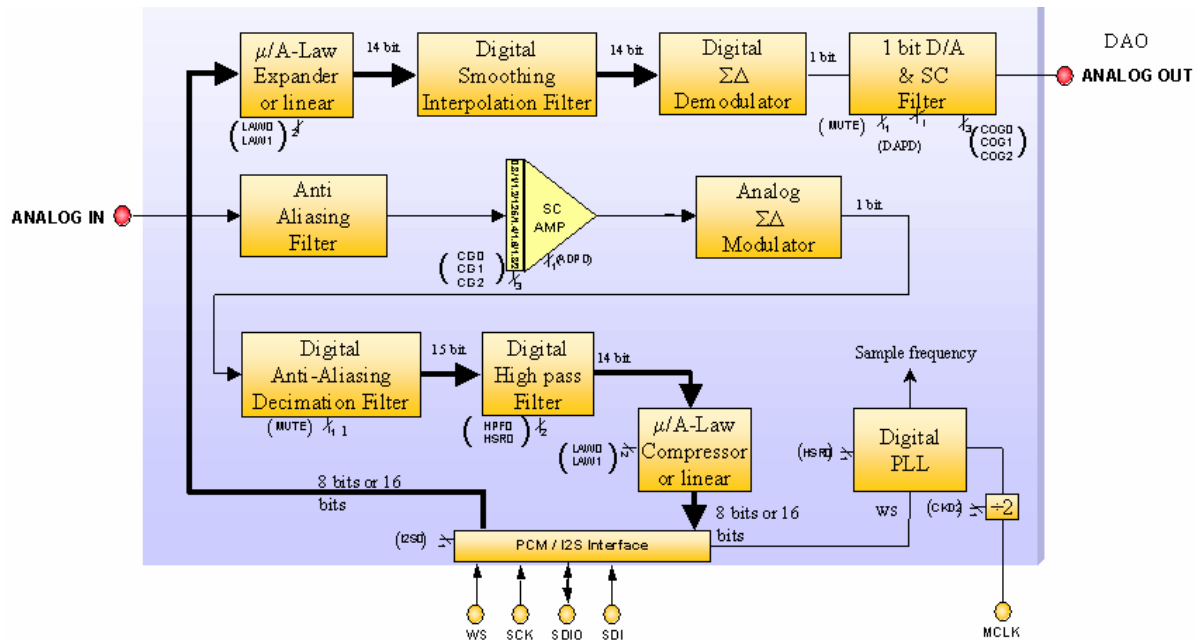
## 7.2. CODEC

The CODEC built into the ISD5216 supports both the I<sup>2</sup>S and PCM digital interface using  $\mu$ -Law and A-Law companding as well as 2's complement and signed magnitude data. The CODEC meets the PCM conformance specification of the G.714 recommendation. Its  $\mu$ -Law and A-Law compander meets the specification of the ITU-T G.711 recommendation.

The CODEC operates in full duplex in PCM mode and half duplex in I<sup>2</sup>S mode. Operating the CODEC requires an external master clock running at 13.824 MHz, 20.48 MHz, 27.648 MHz or 40.96 MHz. This provides a sampling frequency ranging from 8kHz to 48kHz.



The following diagram shows the functional blocks in the CODEC:



### 7.2.1. Analog Input to Digital Output Path

A 200 kHz anti-aliasing filter processes the analog input signal before entering the amplifier for the A/D converter. The gain of this amplifier is adjustable through the configuration registers bits (CIG2 – CIG0) for a gain from 0.80 to 2.00.

The Sigma Delta modulator is a Linear 14 bit  $\Sigma\Delta$  modulator running at a sampling frequency determined by the external clock input and the internal clock dividers (CKD2, CKDV). The standard telecom frequency of 8kHz and digital audio of 44.1kHz and 48 kHz as well as intermediate frequencies as shown in the table on the next page are supported. The A/D converter can be turned off to save power and reduce noise by setting the A/D power down bit (ADPD).

The A/D converter feeds a 3.4 kHz digital anti aliasing filter which can be muted to suppress noise, the mute bit controls both the A/D and D/A filter simultaneously. The following high pass filter is enabled by bit (HPF0) in the configuration register. The High Sampling Rate bit (HSR0) needs to be set to enable operation at 44.1kHz – 48 kHz.

The digital audio signal can be companded using  $\mu$  - Law and A-Law companding or go to the output uncompressed using 2's complement or signed magnitude output selected with bits (LAW1 – LAW0) in the configuration registers.

Finally the digital output interface is selected to be either full-duplex PCM or half duplex I<sup>2</sup>S using the interface selector bit (I<sup>2</sup>S0) in the configuration register. The PCM interface uses the SDIO and SDI pins, the half-duplex I<sup>2</sup>S format uses the SDIO pin as both input and output.



### 7.2.2. Digital Input to Analog Output Path

The digital input interface must be selected to either PCM or I<sup>2</sup>S using the interface selector bit (I<sup>2</sup>S0) in the configuration register. The compression format must also be selected with bits (LAW1 – LAW0) in the configuration registers.

The external clock input signal on pin MCLK and the internal clock dividers must be set to values supporting the selected digital input signal.

The digital smoothing and interpolation filter runs at 3.4 kHz and feeds the  $\Sigma\Delta$  D/A converter that can be switched off to conserve power and reduce noise using the D/A power down bit (DAPD).

The analog output amplifier gain is controlled from configuration registers bits (COG2 – COG0) from -8 dB to +6 dB.

### 7.2.3. CODEC External Clock Configuration

The ISD5216 has two Master Clock configuration bits that allow four possible Master Clock frequencies. Bits CKD2 and CKDV set the Master Clock Division ratios. These are bits D12 and D8 of CFG2, respectively. The combination of these bits, with the sample rate bit HSR0, also set the CODEC sample frequency as shown in the following table.

**Master Clock Possible Settings**

F <sub>MCLK</sub>	HSR0 (D5) (CFG2)	CKD2 (D12) (CFG2)	CKDV (D8) (CFG2)	F <sub>SCODEC</sub>
13.824 MHz	0	0	0	8 kHz
20.48 MHz	0	0	1	11.852 kHz*
27.648 MHz	0	1	0	8 kHz
40.96 MHz	0	1	1	11.852 kHz*
13.824 MHz	1	0	0	32 kHz*
20.48 MHz	1	0	1	44.1 - 48 kHz
27.648 MHz	1	1	0	32 kHz*
40.96 MHz	1	1	1	44.1-48 kHz

\*not tested



#### 7.2.4. ChipCorder Analog Array Sampling Frequency With External Clock

If an external master clock is used, the clock dividers must be set according to the following table to get the filter cut-off frequency and sample rate setup correctly. The duty cycle on the input clock is not critical when CKD2 is set to ONE as the clock is immediately divided by two internally. See the [Analog Structure \(Right Half\)](#) description on page 32.

F <sub>MCLK</sub>	FLD 1	FLD0	CKD2	CKDV	Sample Rate	Filter Knee
13.824 MHz	0	0	0	0	8.0 kHz	3.7 kHz
20.48 MHz	0	0	0	1	8.0 kHz	3.7 kHz
27.648 MHz	0	0	1	0	8.0 kHz	3.7 kHz
40.96 MHz	0	0	1	1	8.0 kHz	3.7 kHz
13.824 MHz	0	1	0	0	6.4 kHz	2.9 kHz
20.48 MHz	0	1	0	1	6.4 kHz	2.9 kHz
27.648 MHz	0	1	1	0	6.4 kHz	2.9 kHz
40.96 MHz	0	1	1	1	6.4 kHz	2.9 kHz
13.824 MHz	1	0	0	0	5.3 kHz	2.5 kHz
20.48 MHz	1	0	0	1	5.3 kHz	2.5 kHz
27.648 MHz	1	0	1	0	5.3 kHz	2.5 kHz
40.96 MHz	1	0	1	1	5.3 kHz	2.5 kHz
13.824 MHz	1	1	0	0	4.0 kHz	1.8 kHz
20.48 MHz	1	1	0	1	4.0 kHz	1.8 kHz
27.648 MHz	1	1	1	0	4.0 kHz	1.8 kHz
40.96 MHz	1	1	1	1	4.0 kHz	1.8 kHz

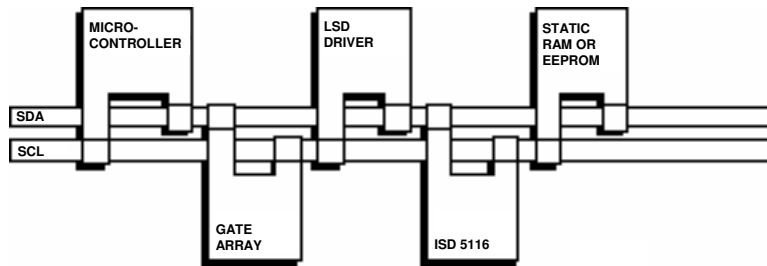


**7.3. I<sup>2</sup>C INTERFACE**

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.

**7.3.1. System configuration**

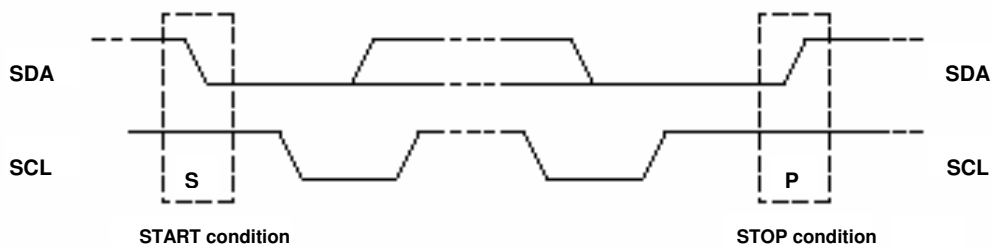
A device generating a message is a 'transmitter'; a device receiving a message is the 'receiver'. The device that controls the message is the 'master' and the devices that are controlled by the master are the 'slaves'.



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

**7.3.2. Start and stop conditions**

Both data and clock lines remain HIGH when the interface bus is not busy. A HIGH-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).



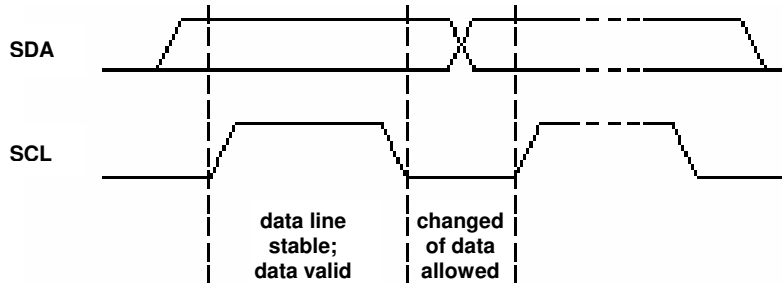
Definition of START and STOP conditions





**7.3.3. 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. The same timing applies to both read and write.

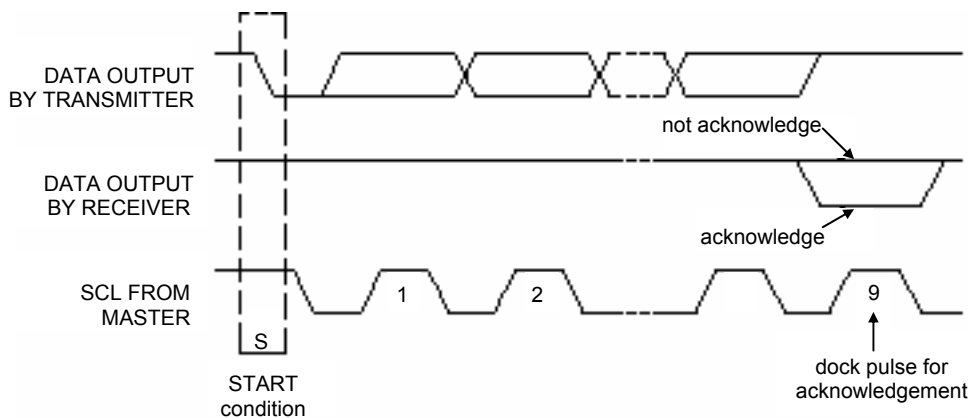


**Bit transfer on the I<sup>2</sup>C-Bus**

**7.3.4. ACKNOWLEDGE**

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**



### 7.3.5. Additional ISD5216 flow control

The I<sup>2</sup>C Interface in the ISD5216 differs from the standard implementation in the way the SCL line is also used for flow control. The ISD5216 will hold the clock line low until it is ready to accept another command/data. The SCL line must be implemented as a bi-directional line like the SDA line.

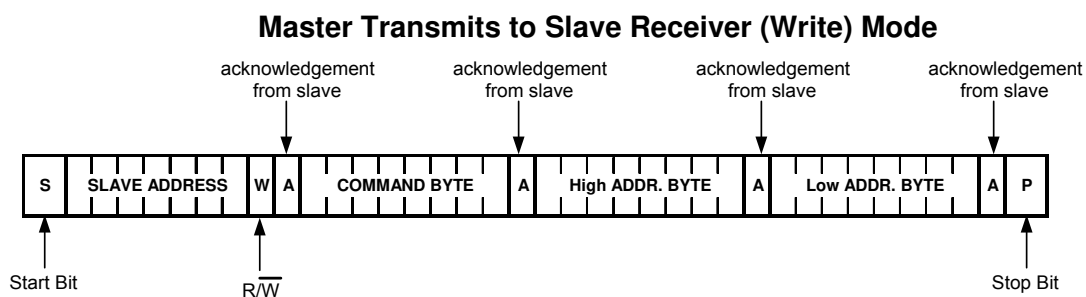
For example, the sequence of sending the slave address will be as follows:

1. Send one byte 10000000 {Slave Address, R/W = 0} 80h.
2. Wait for slave to acknowledge (ACK)
3. Next time the clock is pulled high by the master, wait for SCL to actually go high.

### 7.3.6. I<sup>2</sup>C Protocol Addressing

Since the I<sup>2</sup>C 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.

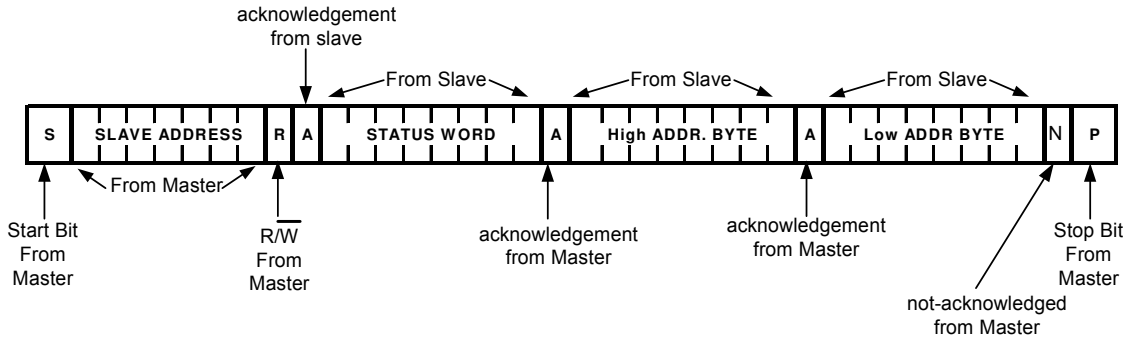
Before any data is transmitted on the I<sup>2</sup>C interface, the current bus master must address the slave it wishes to transfer data to or from. The Slave Address is always sent out as the 1<sup>st</sup> byte following the Start Condition sequence. An example of a Master transmitting an address to a ISD5216 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.



A common procedure in the ISD5216 is the reading of the Status Bytes. The Read Status condition in the ISD5216 is triggered when the Master addresses the chip with its proper Slave Address, immediately followed by the R/W bit set to a "0" 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.



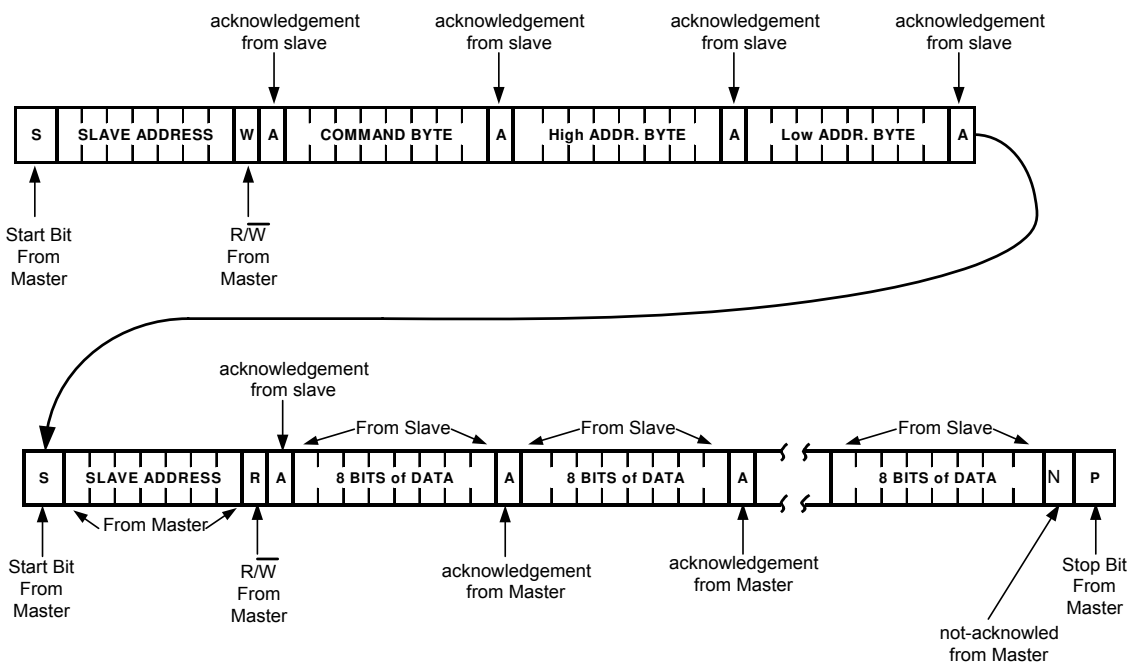
**Master Reads from Slave immediately after first byte (Read Mode)**



Another common operation in the ISD5216 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 ISD5216 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 ISD5216, 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 the section on page 22 of this datasheet.

**Master Reads from the Slave after setting data address in Slave**

**(Write data address, READ Data)**





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

The ISD5216 has a 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 eight possible slave addresses for the ISD5216. To use more than four ISD5216 devices in an application requires some external switching of the I<sup>2</sup>C link.

A1	A0	Slave Address	R/W Bit	HEX Value
0	0	<100 00 00>	0	80
0	1	<100 00 01>	0	82
1	0	<100 00 10>	0	84
1	1	<100 00 11>	0	86
0	0	<100 00 00>	1	81
0	1	<100 00 01>	1	83
1	0	<100 00 10>	1	85
1	1	<100 00 11>	1	87

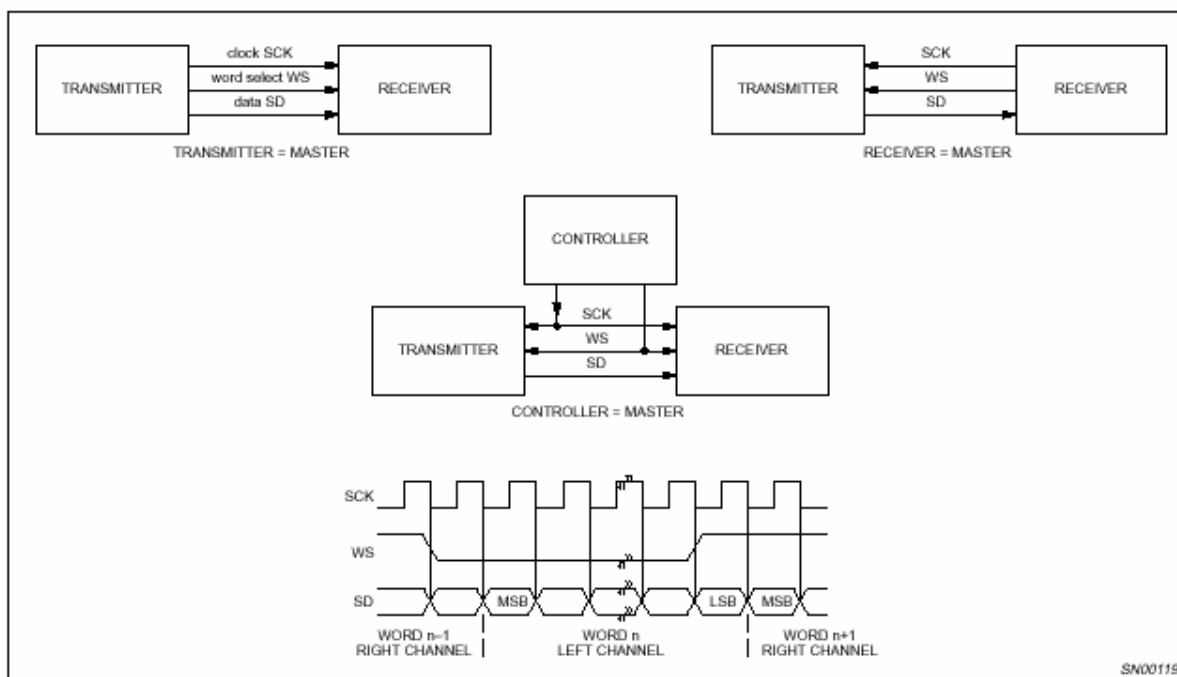


#### 7.4. I2S SERIAL INTERFACE

As shown in the following figure, the bus has three lines:

- continuous serial clock (SCK)
- word select (WS)
- serial data (SDIO) and the device generating SCK and WS is the master.

#### Simple System Configurations and Basic Interface Timing



##### 7.4.1. Serial Data

Serial data is transmitted in two's complement with the MSB first. The MSB is transmitted first because the transmitter and receiver may have different word lengths. It isn't necessary for the transmitter to know how many bits the receiver can handle, nor does the receiver need to know how many bits are being transmitted.

When the system word length is greater than the transmitter word length, the word is truncated (least significant data bits are set to '0') for data transmission. If the receiver is sent more bits than its word length, the bits after the LSB are ignored. On the other hand, if the receiver is sent fewer bits than its word length, the missing bits are set to zero internally. And so, the MSB has a fixed position, whereas the position of the LSB depends on the word length. The transmitter always sends the MSB of the next word one clock period after the WS changes.

Serial data sent by the transmitter may be synchronized with either the trailing (HIGH-to-LOW) or the leading (LOW-to-HIGH) edge of the clock signal. However, the serial data must be latched into the receiver on the leading edge of the serial clock signal, and so there are some restrictions when



transmitting data that is synchronized with the leading edge (see the timing specifications at the back of this data sheet).

Note that the specifications are defined by the transmitter speed. The specification of the receiver has to be able to match the performance of the transmitter.

#### 7.4.2. Word Select

The word select line indicates the channel being transmitted:

- WS = 0; channel 1 (left)
- WS = 1; channel 2 (right)

WS may change either on a trailing or leading edge of the serial clock, but it doesn't need to be symmetrical. In the slave, this signal is latched on the leading edge of the clock signal. The WS line changes one clock period before the MSB is transmitted. This allows the slave transmitter to derive synchronous timing of the serial data that will be set up for transmission. Furthermore, it enables the receiver to store the previous word and clear the input for the next word (see figure [Timing for I<sup>2</sup>S Transmitter](#) on previous page.)

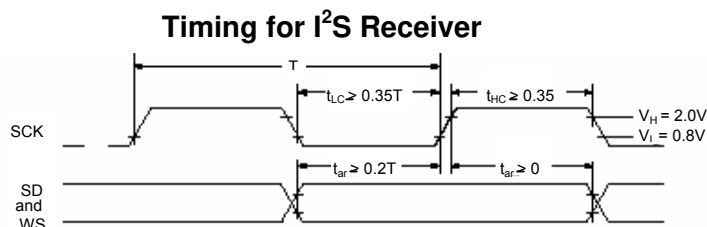
#### 7.4.3. Timing

In the I<sup>2</sup>S format, any device can act as the system master by providing the necessary clock signals. A slave will usually derive its internal clock signal from an external clock input. This means, taking into account the propagation delays between master clock and the data and/or word-select signals, the total delay is simply the sum of:

- the delay between the external (master) clock and the slave's internal clock; and
- the delay between the internal clock and the data and/or word-select signals.

For data and word-select inputs, the external to internal clock delay is of no consequence because it only lengthens the effective set-up time (see figure [Timing for I<sup>2</sup>S Transmitter](#) on previous page.) The major part of the time margin is to accommodate the difference between the propagation delay of the transmitter, and the time required to set up the receiver.

All timing requirements are specified relative to the clock period or to the minimum allowed clock period of a device. This means that higher data rates can be used in the future.



T = clock period

$T_R$  = minimum allowed clock period for transmitter

$T > T_R$

Note that the specifications are defined by the transmitter speed. The specification of the receiver has to be able to match the performance of the transmitter.



## 7.5. CONTROL REGISTERS

The ISD5216 is controlled by loading commands to, or reading commands 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.

### 7.5.1. Command Byte

Control of the ISD5216 is implemented through an 8-bit command byte that is sent after the 7-bit device address and the 1-bit Read/Write selection bit. The 8 bits are:

- Global power up bit (PU)
- 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

C7	C6	C5	C4	C3	C2	C1	C0
PU	DAB	FN2	FN1	FN0	RG2	RG1	RG0
	Function Bits				Register Bits		



### 7.5.2. 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. Note that not all decode combinations are currently used; they are reserved for future use. Out of 16 possible codes, the ISD5216 uses 7 for normal operation. The other 9 are No Ops.

Command 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)

### 7.5.3. 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. [The remaining three codes are No Ops.]

RG2	RG1	RG0	Function
C2	C1	C0	
0	0	0	No action
0	0	1	Load Address
0	1	0	Load CFG0
0	1	1	Load CFG1
1	0	1	Load CFG2





## 7.5.4. OPCODE Command Byte Table

OPCODE	HEX	Pwr	Function Bits				Register Bits		
		PU	DA B	FN 2	FN 1	FN 0	RG 2	RG 1	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 ADDRESS	81	1	0	0	0	0	0	0	1
LOAD CFG0	82	1	0	0	0	0	0	1	0
LOAD CFG1	83	1	0	0	0	0	0	1	1
LOAD CFG2	85	1	0	0	0	0	1	0	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
ERASE DIGITAL PAGE	D0	1	1	0	1	0	0	0	0
ERASE DIGITAL PAGE @ ADDR	D1	1	1	0	1	0	0	0	1
WRITE DIGITAL	C8	1	1	0	0	1	0	0	0
WRITE DIGITAL @ ADDR	C9	1	1	0	0	1	0	0	1
READ DIGITAL	E0	1	1	1	0	0	0	0	0
READ DIGITAL @ ADDR	E1	1	1	1	0	0	0	0	1
READ STATUS REGISTER	N/A	N/A	N/A	N/A	N/A	N/A	N/A	N/A	N/A



### 7.5.5. Power-up

The ISD5216 must be powered up before sending any other commands. Wait for T<sub>pu</sub>d time before sending the next command.

### 7.5.6. Read Status

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 is the upper address byte and the last is the lower address byte. The status register is one byte long and its bit function is:

BIT#	NAME	FUNCTION
7	EOM	Indicates whether an EOM interrupt has occurred.
6	OVF	Indicates whether an overflow interrupt has occurred.
5	READY	Indicates the internal status of the device – if READY is LOW no new commands should be sent to device.
4	PD	Device is powered down if PD is HIGH.
3	PRB	Play/Record mode indicator. HIGH=Play/LOW=Record.
2	DEVICE_ID	An internal device ID. This is 001 for the ISD5216.
1		
0		

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

It is good practice to 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, the controller does not have to poll as frequently to determine the status of the ISD5216

### 7.5.7. Attaching an Address to a Command

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, as specified by 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>

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