TEXAS INSTRUMENTS

TAS5028

8 Channel Digital Audio PWM Processor

PurePath DIGITAL™

Data Manual

2004

DAV - Digital Audio/Speaker

SLES112

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1 Introduction

The TAS5028 is an eight channel digital pulse width modulator (PWM) that provides both advanced performance and a high level of system integration. The TAS5028 is designed to interface seamlessly with most audio digital signal processors. The TAS5028 automatically adjusts control configurations in response to clock and data rate changes and idle conditions. This enables the TAS5028 to provide an easy to use control interface with relaxed timing requirements.

The TAS5028 can drive eight channels of H-bridge power stages. Texas Instruments H-bridge parts TAS5111, TAS5112, or TAS5182 + FETs are designed to work seamlessly with the TAS5028. The TAS5028 supports both single-ended or bridge tied load configurations. The TAS5028 also provides a high performance differential output to drive an external differential input analog headphone amplifier (such as the TPA112).

The TAS5028's uses an AD modulation operating at a 384-kHz switching rate for 48-, 96-, and 192-kHz data. The 8x oversampling combined with the 5th order noise shaper provides a broad flat noise floor and excellent dynamic range from 20 Hz to 20 kHz.

The TAS5028 is clock slave only device. The TAS5028 receives MCLK, SCLK and LRCLK from other system components. The TAS5028 accepts master clock rates of 128, 192, 256, 384, 512, and 768 Fs. The TAS5028 accepts a 64-Fs bit clock.

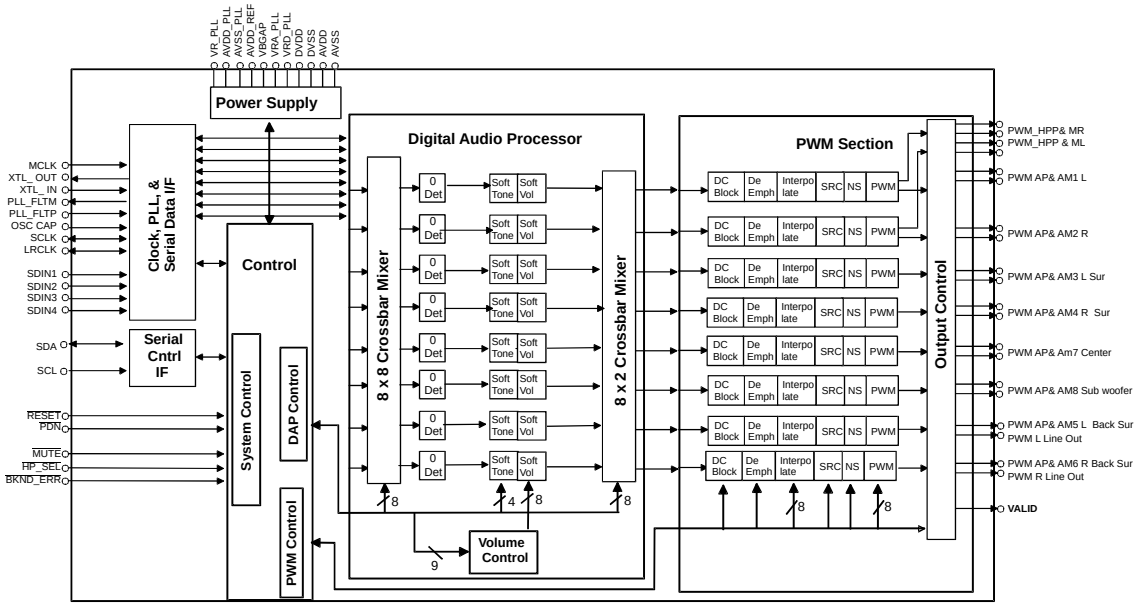


Figure 1-1. TAS5028 Functional Structure

1.1 TAS5028 System Diagrams

Typical applications for the TAS5028 are 6- to 8-channel audio systems such as DVD receiver or AV receiver. Figure 1-2 shows the basic system diagram of the DVD receiver.

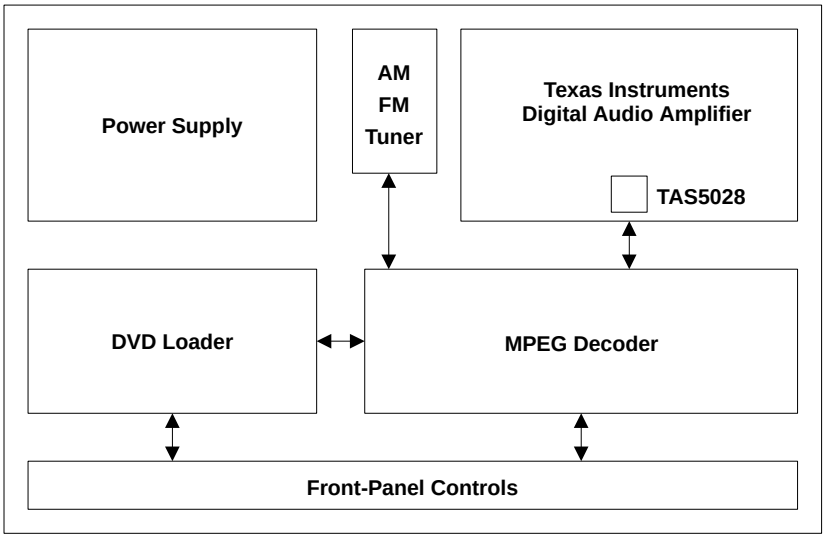


Figure 1-2. Typical TAS5028 Application (DVD Receiver)

Figure 1-3 shows the recommended channel configuration when using the TAS5028 with the TAS5121 power stage. Note that each channel is normally dedicated to a particular function.

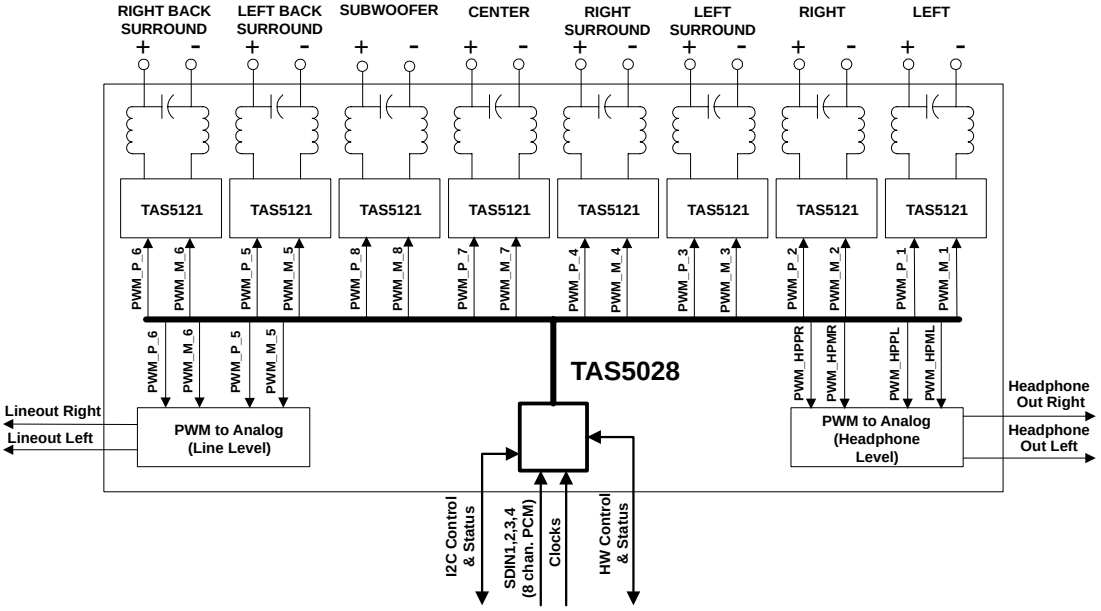


Figure 1-3. Recommended TAS5028 + TAS5121 Channel Configuration

1.2 TAS5028 Features

1.2.1 Audio Input / Output

- Automatic Master Clock Rate and Data Sample Rate Detection
- Eight Serial Audio Input Channels
- Eight PWM Audio Output Channels Configurable as Six Channels With Stereo Line Out or Eight Channels
- Line Output is a PWM Output to Drive an External Differential Input Operational Amplifier
- Headphone PWM Output to Drive an External Differential Amplifier Like the TPA112
- PWM Outputs Support Single Ended and Bridge Tied Loads
- 32-, 38-, 44.1-, 48-, 88.2-, 96-, 176.4-, and 192-kHz Sampling Rates
- Data Formats: 16-, 20-, or 24-bit input Data Left, Right and I^2S,
- 64 x Fs Bit Clock Rate
- 128, 192, 256, 384, 512, and 768 x Fs Master Clock Rates (Up to a Maximum of 50 MHz)

1.2.2 Audio Processing

- 48-Bit Processing Architecture With 76 bits of Precision for Most Audio Processing Features
- Volume Control Range +36 dB to – 127 dB
  - Master Volume Control Range of +18 dB to –100 dB
  - Eight Individual Channel Volume Control Range of +18-dB to -127-dB
- Programmable Soft Volume and Mute Update Rates
- Four Bass and Treble Tone Controls with ±18-dB Range, Selectable Corner Frequencies, and 2^{nd} Order Slopes
  - L, R, and C
  - LS, RS
  - LR, RR
  - Sub
- Full 8x8 Input Crossbar Mixer. Each Signal Processing Channel Input Can Be Any Ratio of the Eight Input Channels
- 8x2 Output Mixer – Channels 1-6. Each Output Can Be Any Ratio of Any Two Signal Processed Channels
- 8x3 Output mixer – Channels 7 and 8. Each Output can be Any Ratio of Any Three Signal Processed Channels
- Three Coefficient Sets Stored on the Device Can be Selected Manually or Automatically (Based on Specific Data Rates)
- DC Blocking Filters
- Able to Support a Variety of Bass Management Algorithms

1.2.3 PWM Processing

- 32-Bit Processing PWM Architecture With 40 Bits of Precision
- 8x Oversampling With 5th Order Noise Shaping at 32 – 48 kHz, 4x Oversampling at 88.2 kHz, and 96 kHz and 2x Oversampling at 176.4 kHz and 192 kHz
- >102-dB Dynamic Range
- THD+N < 0.1%
- 20 – 20-kHz Flat Noise Floor for 44.1-, 48-, 88.2-, 96-, 176.4-, and 192-kHz Data Rates
- Digital De-emphasis for 32-, 44.1-, and 48-kHz Data Rates
- Flexible Automute Logic With Programmable Threshold and Duration for Noise Free Operation
- Intelligent AM Interference Avoidance System Provides Clear AM Reception
- Adjustable Modulation Limit

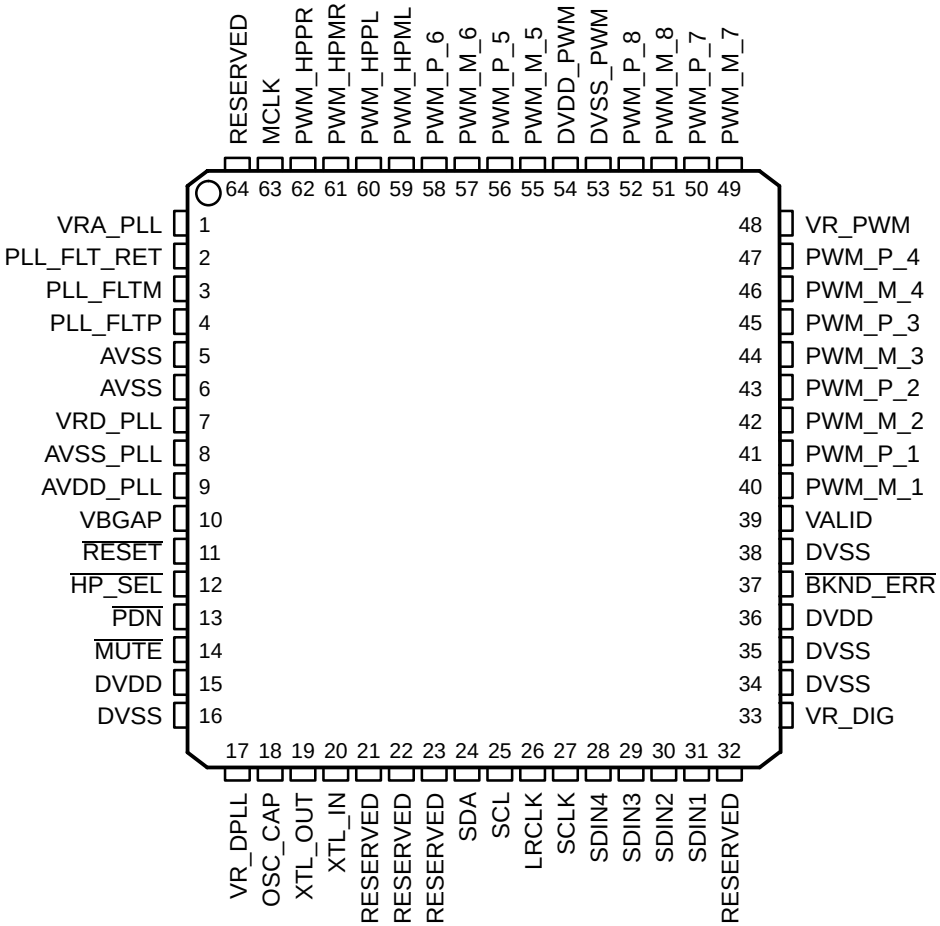
1.2.4 General Features

- Automated Operation With an Easy to Use Control Interface
- I^2C Serial Control Slave Interface
- Integrated AM Interference Avoidance Circuitry
- Single 3.3-V Power Supply
- 64-Pin TQFP Package
- 5-V Tolerant Inputs

1.3 Physical Characteristics

1.3.1 Terminal Assignments

TQFP PACKAGE
(TOP VIEW)



1.3.2 Ordering Information

TA	PLASTIC 64-PIN PQFP (PN)
0°C to 70°C	TAS5028PAG

1.3.3 Terminal Descriptions

TERMINAL NO.	TERMINAL NAME	I/O	5-V TOLERANT	TERMINATION	DESCRIPTION
1	VRA_PLL				Voltage reference for PLL analog supply 1.8 V. A pin-out of the internally regulated 1.8-V power used by PLL logic. A 0.1-µF low ESR capacitor should be connected between this terminal and AVSS_PLL. This terminal must not be used to power external devices.
2	PLL_FLT_RET	AO			PLL external filter return
3	PLL_FLTM	AO			PLL negative input. Connected to PLL_FLT_RTN via an RC network
4	PLL_FLTP	AI			PLL positive input. Connected to PLL_FLT_RTN via an RC network
5	AVSS	P			Analog ground
6	AVSS	P			Analog ground

TERMINAL NO.	TERMINAL NAME	I/O	5-V TOLERANT	TERMINATION	DESCRIPTION
7	VRD_PLL	P			Voltage reference for PLL digital supply 1.8 V. A pin-out of the internally regulated 1.8-V power used by PLL logic. A 0.1-µF low ESR capacitor should be connected between this terminal and AVSS_PLL. This terminal must not be used to power external devices.
8	AVSS_PLL	P			Analog ground for PLL. This terminal should reference the same ground as power terminal DVSS, but to achieve low PLL jitter; ground noise at this terminal must be minimized. The availability of the AVSS terminal allows a designer to use optimizing techniques such as star ground connections, separate ground planes, or other quiet ground distribution techniques to achieve a quiet ground reference at this terminal.
9	AVDD_PLL	P			3.3-V analog power supply for PLL This terminal can be connected to the same power source used to drive power terminal DVSS, but to achieve low PLL jitter, this terminal should be bypassed to AVSS_PLL with a 0.1-µF low-ESR capacitor.
10	VBGAP	P			Band gap voltage reference. A pin-out of the internally regulated 1.2-V reference. Typically has a 1-nF low ESR capacitor between VBGAP and AVSS_PLL. This terminal must not be used to power external devices.
11	$\overline{\text{RESET}}$	DI	5 V	Pull up	System reset input, active low. A system reset is generated by applying a logic low to this terminal. $\overline{\text{RESET}}$ is an asynchronous control signal that restores the TAS5028 to its default conditions, sets the valid output low, and places the PWM in the hard mute (M) state. Master volume is immediately set to full attenuation. Upon the release of $\overline{\text{RESET}}$, if $\overline{\text{PDN}}$ is high, the system performs a 4-5 ms. device initialization and set the volume at mute.
12	$\overline{\text{HP_SEL}}$	DI	5 V	Pull up	Headphone in/out selector. When a logic low is applied, the headphone is selected (speakers are off). When a logic high is applied, speakers are selected – headphone is off.
13	$\overline{\text{PDN}}$	DI	5 V	Pull up	Power down, active low. PDN powers down all logic and stops all clocks whenever a logic low is applied. The internal parameters are preserved through a power down cycle, as long as a RESET is not active. The duration for system recovery from power down is 100 ms.
14	$\overline{\text{MUTE}}$	DI	5 V	Pull up	Soft mute of outputs, active low (Muted signal = a logic low, normal operation = a logic high) The mute control provides a noiseless volume ramp to silence. Releasing mute provides a noiseless ramp to previous volume.
15	DVDD	P			Digital power 3.3-V supply for digital core and most of I/O buffers
16	DVSS	P			Digital ground for digital core and most of I/O buffers
17	VR_DPLL	P			Voltage reference for digital PLL supply 1.8 V. A pin-out of the internally regulated 1.8-V power used by digital PLL logic. A 0.1-µF low ESR capacitor should be connected between this terminal and DVSS_CORE. This terminal must not be used to power external devices.
18	OSC_CAP	AO			Oscillator capacitor
19	XTL_OUT	AO			XTL_OUT and XTL_IN are the only LVCMOS terminals on the device. They provide a reference clock for the TAS5028 via use of an external fundamental mode crystal. XTL_OUT is the 1.8-V output drive to the crystal. See Note 4 for the recommended crystal type.
20	XTL_IN	AI			XTL_OUT and XTL_IN are the only LVCMOS terminals on the device. They provide a reference clock for the TAS5028 via use of an external fundamental mode crystal. XTL_IN is the 1.8-V input port for the oscillator circuit. See Note 4 for the recommended crystal type.
21	RESERVED				Connect to digital ground
22	RESERVED				Connect to digital ground
23	RESERVED				Connect to digital ground
24	SDA	DIO	5 V		I^2C serial control data interface input / output
25	SCL	DI	5 V		I^2C serial control clock input output
26	LRCLK	DI	5 V		Serial audio data left / right clock (sampling rate clock)
27	SCLK	DI	5 V		Serial audio data clock (shift clock) SCLKIN is the serial audio port (SAP) input data bit clock that is supplied to the serial bit clock to other I^2S bus.

TERMINAL NO.	TERMINAL NAME	I/O	5-V TOLERANT	TERMIN-ATION	DESCRIPTION
28	SDIN4	DI	5 V	Pulldown	Serial audio data 4 input is one of the serial data input ports. SDIN4 supports four discrete (stereo) data formats and is capable of inputting data at 64 Fs.
29	SDIN3	DI	5 V	Pulldown	Serial audio data 3 input is one of the serial data input ports. SDIN3 supports four discrete (stereo) data formats and is capable of inputting data at 64 Fs.
30	SDIN2	DI	5 V	Pulldown	Serial audio data 2 input is one of the serial data input ports. SDIN2 supports four discrete (stereo) data formats and is capable of inputting data at 64 Fs.
31	SDIN1	DI	5 V	Pulldown	Serial audio data 1 input is one of the serial data input ports. SDIN1 supports four discrete (stereo) data formats and is capable of inputting data at 64 Fs.
32	RESERVED	O			
33	VR_DIG	P			Voltage reference for digital core supply 1.8 V. A pin-out of the internally regulated 1.8-V power used by digital core logic. A 0.47-µF low ESR capacitor should be connected between this terminal and DVSS. This terminal must not be used to power external devices
34	DVSS	P			Digital ground
35	DVSS	P			Digital ground
36	DVDD	P			3.3-V digital power supply
37	$\overline{\text{BKND_ERR}}$	DI		Pull up	Active low. A backend error sequence is generated by applying logic low to this terminal. The $\overline{\text{BKND_ERR}}$ results in all system parameters unaffected, while all H-bridge drive signals going to a hard mute (M) state.
38	DVSS	P			Digital ground
39	VALID	DO			Output indicating validity of PWM outputs active high
40	PWM_M_1	DO			PWM 1 output (differential -)
41	PWM_P_1	DO			PWM 1 output (differential +)
42	PWM_M_2	DO			PWM 2 output (differential -)
43	PWM_P_2	DO			PWM 2 output (differential +)
44	PWM_M_3	DO			PWM 3 output (differential -)
45	PWM_P_3	DO			PWM 3 output (differential +)
46	PWM_M_4	DO			PWM 4 output (differential -)
47	PWM_P_4	DO			PWM 4 output (differential +)
48	VR_PWM	P			Voltage reference for digital PWM core supply 1.8 V. A pin-out of the internally regulated 1.8-V power used by digital PWM core logic. A 0.1-µF low ESR capacitor should be connected between this terminal and DVSS_PWM. This terminal must not be used to power external devices.
49	PWM_M_7	DO			PWM 7 (Line out L) output (differential -)
50	PWM_P_7	DO			PWM 7 (Line out L) output (differential +)
51	PWM_M_8	DO			PWM 8 (Line out R) output (differential -)
52	PWM_P_8	DO			PWM 8 (Line out R) output (differential +)
53	DVSS_PWM	P			Digital ground for PWM
54	DVDD_PWM	P			3.3-V digital power supply for PWM
55	PWM_M_5	DO			PWM 5 output (differential -)
56	PWM_P_5	DO			PWM 5 output (differential +)
57	PWM_M_6	DO			PWM 6 output (differential -)
58	PWM_P_6	DO			PWM 6 output (differential +)
59	PWM_HPML	DO			PWM left channel headphone (differential -)
60	PWM_HPPL	DO			PWM left channel headphone (differential +)
61	PWM_HPMR	DO			PWM right channel headphone (differential -)
62	PWM_HPPR	DO			PWM right channel headphone (differential +)

TERMINAL NO.	TERMINAL NAME	I/O	5-V TOLERANT	TERMIN-ATION	DESCRIPTION
63	MCLK	DI	5 V	Pulldown	MCLK is a 3.3-V clock master clock input. The input frequency of this clock can range from 4 MHz to 50 MHz.
64	RESERVED				Connect to digital ground

NOTES: 1. Type: A = analog; D = 3.3-V digital; P = power / ground / decoupling; I = input; O = output
2. All pullups are 200-μA weak pullups and all pulldowns are 200-μA weak pull downs. The pullups and pulldowns are included to assure proper input logic levels if the terminals are left unconnected (pullups => logic 1 input; pulldowns => logic 0 input). Devices that drive inputs with pull ups must be able to sink 200 μA, while maintaining a logic 0 drive level. Devices that drive inputs with pulldowns must be able to source 200 μA, while maintaining a logic '1' drive level.
3. If desired, low ESR capacitance values can be implemented by paralleling two or more ceramic capacitors of equal value. Paralleling capacitors of equal value provide an extended high frequency supply decoupling. This approach avoids the potential of producing parallel resonance circuits that have been observed when paralleling capacitors of different values.
4. 13.5-MHz crystal (HCM49)

1.4 TAS5028 Functional Description

Figure 1-4 shows the TAS5028 functional structure. The next sections describe the TAS5028 functional blocks:

- Power Supply
- Clock, PLL, and Serial Data Interface
- Serial Control Interface
- Device Control
- Digital Audio Processor (DAP)
- Pulse Width Modulation (PWM) Processor

1.4.1 Power Supply

The power supply section contains supply regulators that provide analog and digital regulated power for various sections of the TAS5028. The analog supply supports the analog PLL, while digital supplies support the digital PLL, the digital audio processor (DAP), the pulse width modulator (PWM), and the output control (reclocker). The regulators can also be turned off when terminals $\overline{\text{RESET}}$ and $\overline{\text{PDN}}$ are both low.

1.4.2 Clock, PLL, and Serial Data Interface

The TAS5028 is a clock slave only device and it requires the use of an external 13.5 MHz crystal. It accepts MCLK, SCLK, and LRCLK as inputs only.

The TAS5028 uses the external crystal to provide a time base for:

- Continuous data and clock error detection and management
- Automatic data rate detection and configuration
- Automatic MCLK rate detection and configuration (automatic bank switching)
- Supporting I^2C operation/ communication while MCLK is absent

The TAS5028 automatically handles clock errors, data rate changes, and master clock frequency changes without requiring intervention from an external system controller. This feature significantly reduces system complexity and design.

1.4.2.1 Serial Audio Interface

The TAS5028 operates as a slave only / receive only serial data interface in all modes. The TAS5028 has four PCM serial data interfaces to permit eight channels of digital data to be received though the SDIN1, SDIN2, SDIN3, and SDIN4 inputs. The serial audio data is in MSB first, two's complement format.

The serial data input interface of the TAS5028 can be configured in right justified, I^2S, or left-justified modes. The serial data interface format is specified using the I^2C data interface control register. The supported formats and word lengths are shown in Table 1-1.

Table 1-1. Serial Data Formats

RECEIVE SERIAL DATA INTERFACE FORMAT	WORD LENGTHS
Right justified	16
Right justified	20
Right justified	24
I2S	16
I2S	20
I2S	24
Left Justified	16
Left Justified	20
Left Justified	24

Serial data is input on SDIN1, SDIN2, SDIN3, and SDIN4. The TAS5028 accepts 32-, 38-, 44.1-, 48-, 88.2-, 96-, 176.4-, and 192-kHz serial data in 16-, 20-, or 24-bit data in left, right, and I^2S serial data formats using a 64-Fs SCLK clock and a 128, 192, 256, 384, 512, or 768 x Fs MCLK rates (up to a maximum of 50 MHz). The parameters of this clock and serial data interface are I^2C configurable.

1.4.3 I^2C Serial Control Interface

The TAS5028 has an I^2C serial control slave interface (address 0x36) to receive commands from a system controller. The serial control interface supports both normal-speed (100 kHz) and high-speed (400 kHz) operations without wait states. Since the TAS5028 has a crystal time base, this interface operates even when MCLK is absent.

The serial control interface supports both single byte and multi-byte read / write operations for status registers and the general control registers associated with the PWM. However, for the DAP data processing registers, the serial control interface also supports multiple byte (4 byte) write operations.

The I^2C supports a special mode which permits I^2C write operations to be broken up into multiple data write operations that are multiples of 4 data bytes. These are 6 byte, 10 byte, 14 byte, 18 byte ... etc write operations that are composed of a device address, read/write bit, and subaddress and any multiple of 4 bytes of data. This permits the system to incrementally write large register values without blocking other I^2C transactions. In order to use this feature, the first chunk of data is written to the target I^2C address and each chunk of subsequent data is written to a special append register (0xFE), until all the data is written and a stop bit is sent. An incremental read operation is not supported.

1.4.4 Device Control

The TAS5028 control section provides the control and sequencing for the TAS5028. The device control provides both high and low level control for the serial control interface, clock and serial data interfaces, digital audio processor, and pulse width modulator sections.

1.4.5 Digital Audio Processor (DAP)

The DAP arithmetic unit is used to implement all audio processing functions – soft volume, bass and treble processing, and input and output mixing. Figure 1-6 shows the TAS5028 DAP architecture.

The DAP accepts 24-bit data signal from the serial data interface and outputs 32-bit data to the PWM section. The DAP supports two configurations, one for 32-kHz – 96-kHz data and one for 176.4-kHz to 192-kHz data.

1.4.5.1 TAS5028 Audio Processing Configurations

The 32 - 96 kHz configuration supports eight channels of data processing that can be configured as eight channels or six channels with two channels for separate stereo line outputs.

The 176.4 - 192 kHz configuration supports three channels of signal processing with five channels passed though (or derived from the three processed channels).

To efficiently support the processing requirements of both multi-channel 32 – 96-kHz data and the two channel 176.4 and 192-kHz data, the TAS5028 supports separate audio processing features for 32 –96-kHz data rates and for 176.4 and 192 kHz. See Table 2 for a summary of TAS5028 processing feature sets.

1.4.5.2 TAS5028 Audio Signal Processing Functions

The DAP provides 7 primary signal processing functions.

1. The data processing input has a full 8x8 input crossbar mixer. This enables each input to be any ratio of the eight input channels.
2. Two I^2C programmable threshold detectors in each channel support auto mute.
3. Four soft bass and treble tone controls with ±18 dB range, programmable corner frequencies, and 2nd order slopes. In 8-channel mode, bass and treble controls are normally configured as follows:
   - Bass and Treble 1: Channel 1 (Left), Channel 2 (Right), and Channel 7 (Center)
   - Bass and Treble 2: Channel 3 (Left Surround) and Channel 4 (Right Surround)
   - Bass and Treble 3: Channel 5 (Left Back Surround) and Channel 6 (Right Back Surround)
   - Bass and Treble 4: Channel 8 (Subwoofer)
4. Individual channel and master volume controls. Each control provides an adjustment range of +18 dB to –127 dB. This permits a total volume device control range of +36 dB to –127 dB plus mute. The master volume control can be configured to control six or eight channels. The DAP soft volume and mute update interval is I^2C programmable. The update is performed at a fixed rate regardless of the sample rate.
5. 8x2 output mixer (channels 1-6). Each output can be any ratio of any two signal processed channels.
6. 8x3 output mixer (channels 7 and 8). Each output can be any ratio of any three signal processed channels.
7. The DAP maintains three sets of coefficient banks that are used to maintain separate sets of sample rate dependent parameters for the tone controls in RAM. These can be set to be automatically selected for one or more data sample rates or can be manually selected under I^2C program control. This feature enables coefficients for different sample rates to be stored in the TAS5028 and then select when needed.

Table 1-2. TAS5028 Audio Processing Feature Sets

FEATURE	32 - 96 kHz 8 CHANNEL FEATURE SET	32 - 96 kHz 6 + 2 LINE OUT FEATURE SET	176.4- AND 192-kHz FEATURE SET
Signal processing channels	8	6 + 2	3
Pass through channels	N/A		5
Master volume	1 for eight channels	1 for six channels	1 for three channels
Individual channel volume controls	8		3
Bass and treble tone controls	Four Bass and Treble tone controls with ±18 dB range, programmable corner frequencies, and 2nd order slopes L, R and C (Ch 1, 2, and 7) LS, RS (Ch 3 and 4) LBS, RBS (Ch 5 and 6) Sub (Ch 8)	Four Bass and Treble tone controls with ±18 dB range, programmable corner frequencies, and 2nd order slopes L, R and C (Ch 1, 2, and 7) LS, RS (Ch 3 and 4) Sub, (Ch 8) Line L and R (Ch 5 and 6)	Two Bass and Treble tone controls with ±18 dB range, programmable corner frequencies, and 2nd order slopes L and R (Ch 1 and 2) Sub (Ch 8)
Input output mapping/mixing	Each of the eight signal-processing channels input can be any ratio of the eight input channels. Each of the eight outputs can be any ratio of any two processed channels.		Each of the three signal-processing channels or the five pass-though channels inputs can be any ratio of the eight input channels. Each of the eight outputs can be any ratio of any of the three processed or five bypass channels.
DC blocking filters (Implemented in PWM Section)	Eight channels		
Digital de-emphasis (Implemented in PWM Section)	Eight channels for 32 kHz, 44.1 kHz, and 48 kHz	Six channels for 32 kHz, 44.1 kHz, and 48 kHz	N/A
Number of Coefficient sets Stored	Three additional coefficient sets can be stored in memory		

1.5 TAS5028 DAP Architecture

1.5.1 TAS5028 *DAP Architecture Diagrams*

Figure 1-4 shows the TAS5028 DAP architecture for Fs = 96 kHz. Note the TAS5028 bass management architecture shown in channels 1, 2, 7, and 8. Note that the I^2C registers are shown to help the designer configure the TAS5028.

Figure 1-5 shows the TAS5028 architecture for Fs = 176.4 kHz or Fs = 192 kHz. Note that only channels 1, 2, and 8 contain all the features. Channels 3-7 are pass-through except for master volume control.

Figure 1-6 shows TAS5028 detailed channel processing. The output mixer is 8X2 for channels 1-6 and *X3 for channels 7 and 8.

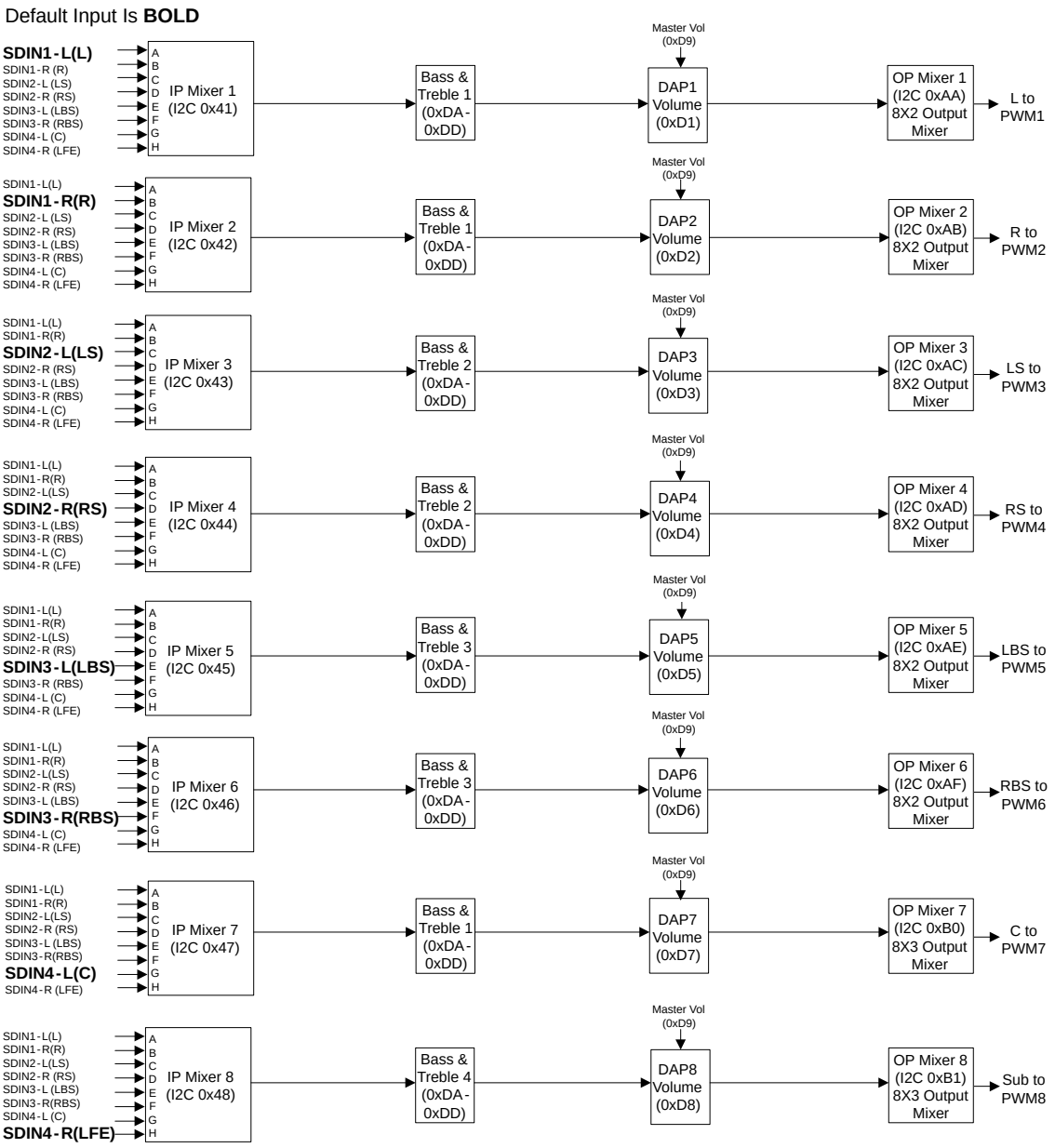


Figure 1-4. TAS5028 DAP Architecture With I²C Registers (Fs ≤ 96 kHz)

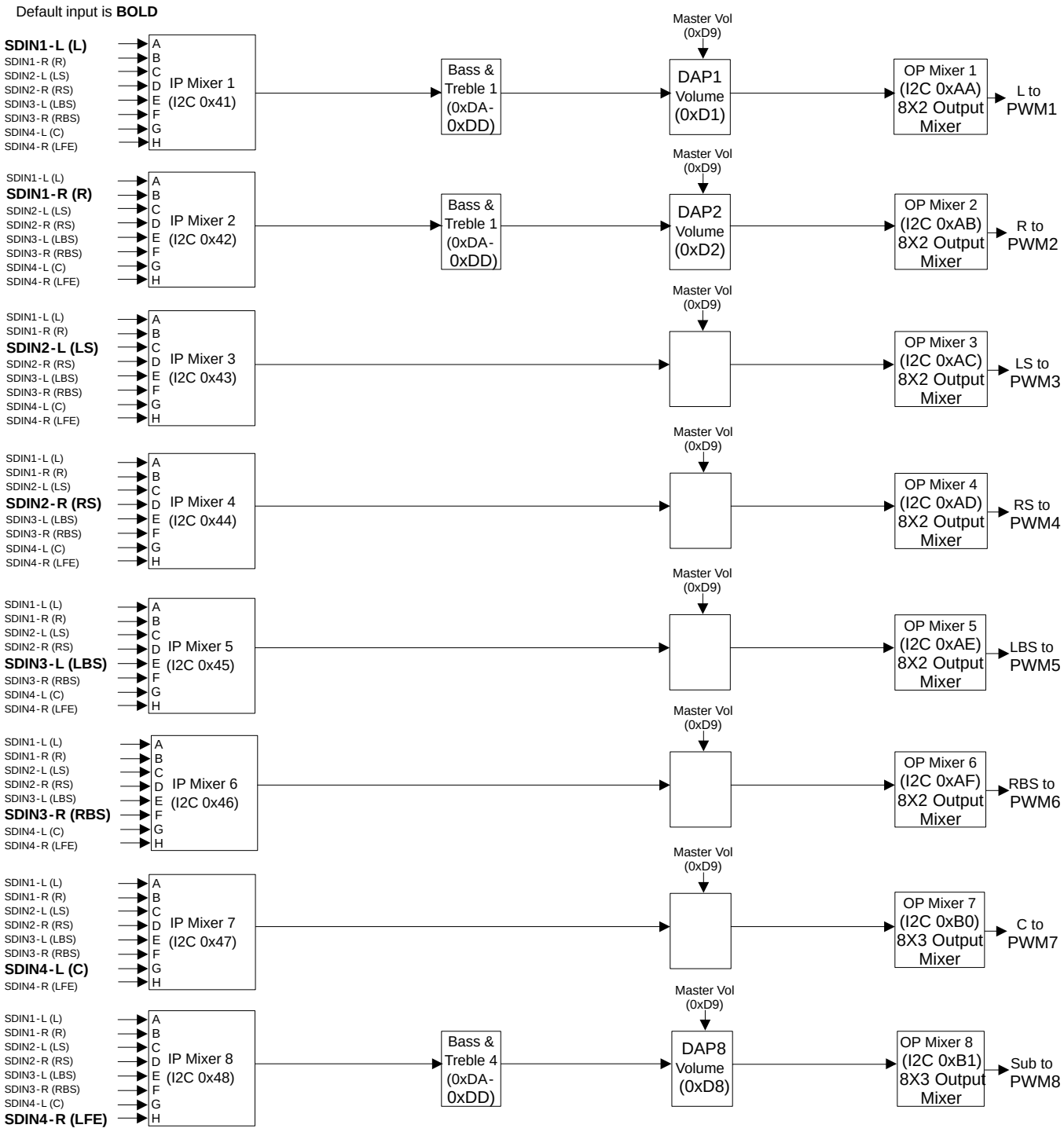


Figure 1-5. TAS5028 Architecture With I²C Registers (Fs = 176.4 kHz or Fs = 192 kHz)

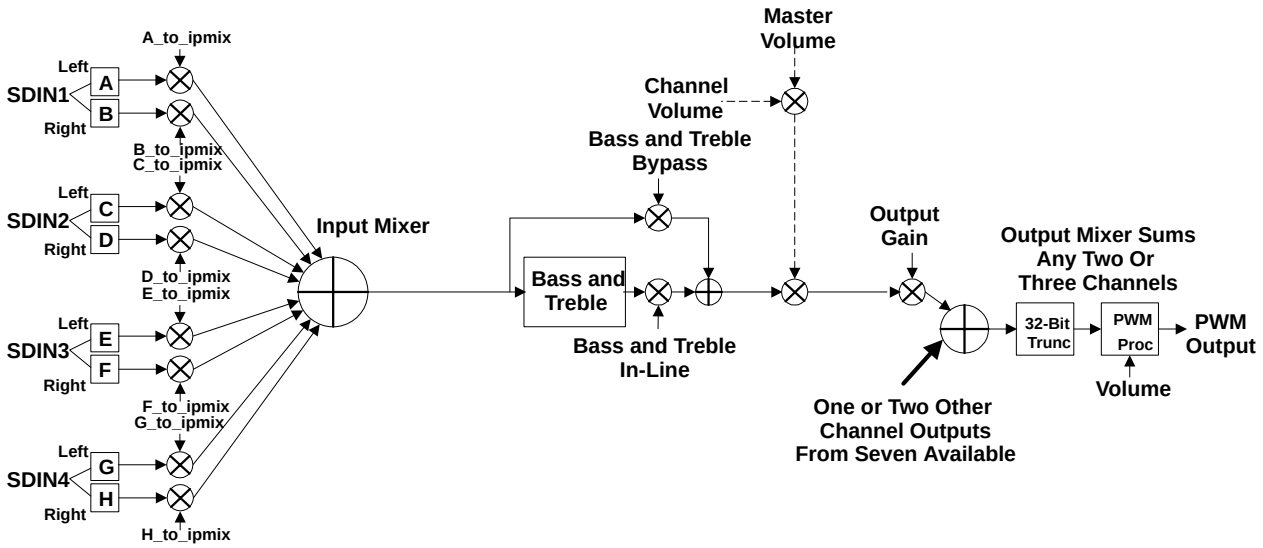


Figure 1-6. TAS5028 Detailed Channel Processing

1.5.2 I²C Coefficient Number Formats

The architecture of the TAS5028 is contained in ROM resources within the TAS5028 and cannot be altered. However, mixer gain, level offset, and filter tap coefficients, which can be entered via the I²C bus interface, provide a user with the flexibility to set the TAS5028 to a configuration that achieves the system level goals.

The firmware is executed in a 48-bit signed fixed-point arithmetic machine. The most significant bit of the 48-bit data path is a sign bit, and the 47 lower bits are data bits. Mixer gain operations are implemented by multiplying a 48-bit signed data value by a 28-bit signed gain coefficient. The 76-bit signed output product is then truncated to a signed 48-bit number. Level offset operations are implemented by adding a 48-bit signed offset coefficient to a 48-bit signed data value. In most cases, if the addition results in overflowing the 48-bit signed number format, saturation logic is used. This means that if the summation results in a positive number that is greater than 0x7FFF_FFFF_FFFF (the spaces are used to ease the reading of the hexadecimal number), the number is set to 0x7FFF_FFFF_FFFF. If the summation results in a negative number that is less than 0x8000_0000_0000 0000, the number is set to 0x8000_0000_0000 0000.

1.5.2.1 28-Bit 5.23 Number Format

All mixer gain coefficients are 28-bit coefficients using a 5.23 number format. Numbers formatted as 5.23 numbers means that there are 5 bits to the left of the decimal point and 23 bits to the right of the decimal point. This is shown in the Figure 1-7.

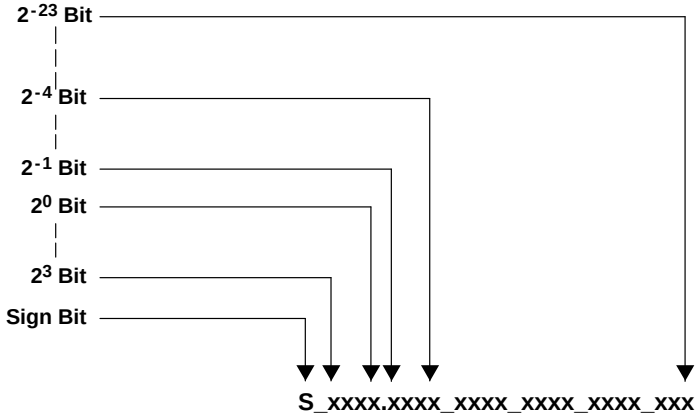


Figure 1-7. 5.23 Format

The decimal value of a 5.23 format number can be found by following the weighting shown in Figure 1-8. If the most significant bit is logic 0, the number is a positive number, and the weighting shown yields the correct number. If the most significant bit is a logic 1, then the number is a negative number. In this case every bit must be inverted, a 1 added to the result, and then the weighting shown in Figure 1-8 applied to obtain the magnitude of the negative number.

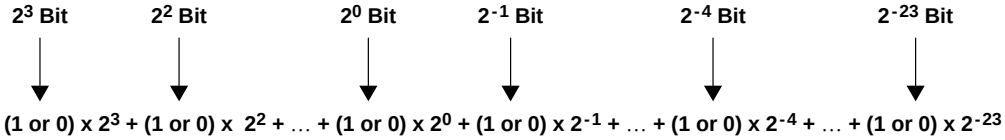
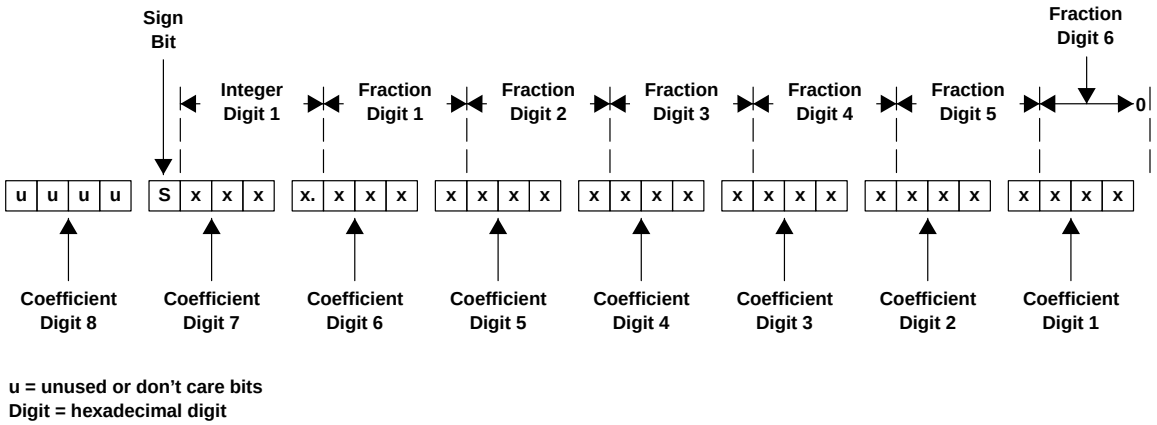


Figure 1-8. Conversion Weighting Factors—5.23 Format to Floating Point

Gain coefficients, entered via the I^2C bus, must be entered as 32-bit binary numbers. The format of the 32-bit number (4-byte or 8-digit hexadecimal number) is shown in Figure 1-9.



u = unused or don't care bits
Digit = hexadecimal digit

Figure 1-9. Alignment of 5.23 Coefficient in 32-Bit I^2C Word

As Figure 1-9 shows, the hex value of the integer part of the gain coefficient cannot be concatenated with the hex value of the fractional part of the gain coefficient to form the 32-bit I^2C coefficient. The reason is that the 28-bit coefficient contains 5 bits of integer, and thus the integer part of the coefficient occupies all of one hex digit and the most significant bit of the second hex digit. In the same way, the fractional part occupies the lower 3 bits of the second hex digit, and then occupies the other five hex digits (with the eighth digit being the zero-valued most significant hex digit).

1.5.2.2 48-Bit 25.23 Number Format

All level adjustment and threshold coefficients are 48-bit coefficients using a 25.23 number format. Numbers formatted as 25.23 numbers means that there are 25 bits to the left of the decimal point and 23 bits to the right of the decimal point. This is shown in Figure 1-10.

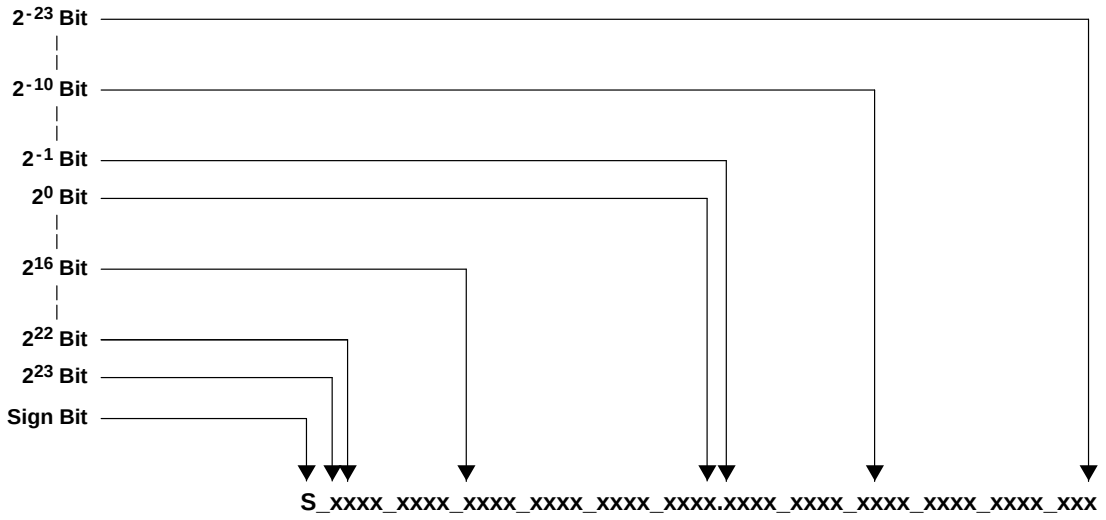


Figure 1-10. 25.23 Format

Figure 1-11 shows the derivation of the decimal value of a 48-bit 25.23 format number.

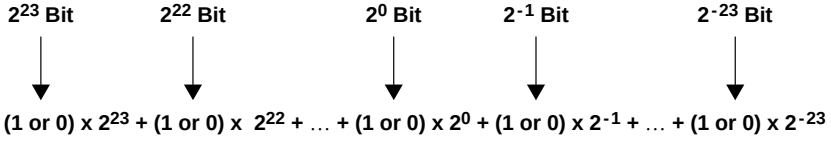


Figure 1-11. Alignment of 5.23 Coefficient in 32-Bit I^{2}C Word

Two 32-bit words must be sent over the I^{2}C bus to download a level or threshold coefficient into the TAS5028. The alignment of the 48-bit, 25.23 formatted coefficient in the 8-byte (two 32-bit words) I^{2}C word is shown in Figure 1-12.

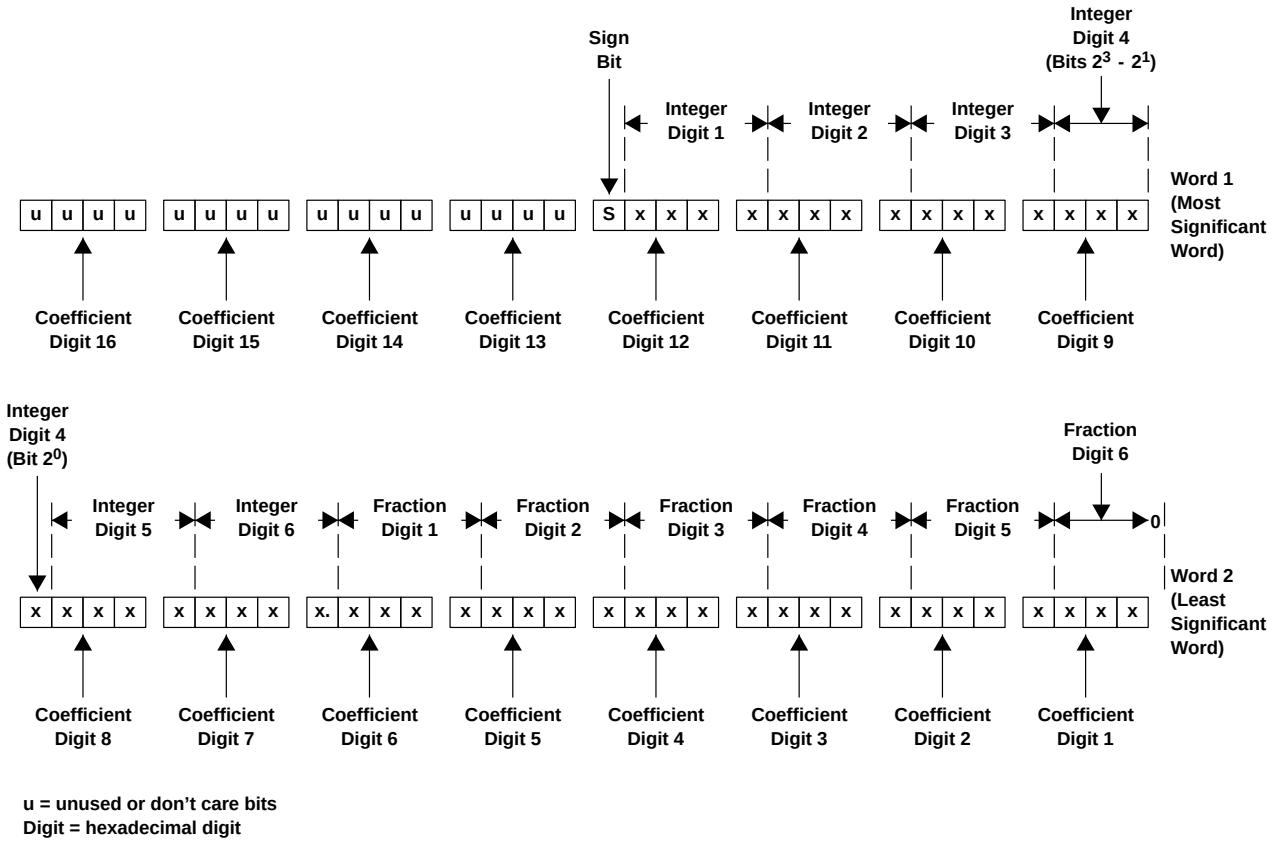


Figure 1-12. Alignment of 25.23 Coefficient in Two 32-Bit I²C Words

1.5.2.3 TAS5028 Audio Processing

The TAS5028 digital audio processing is designed such that noise produced by filter operations is maintained below the smallest signal amplitude of interest, as shown in Figure 1-13. The TAS5028 achieves this by increasing the precision of the signal representation substantially above the number of bits that are absolutely necessary to represent the input signal.

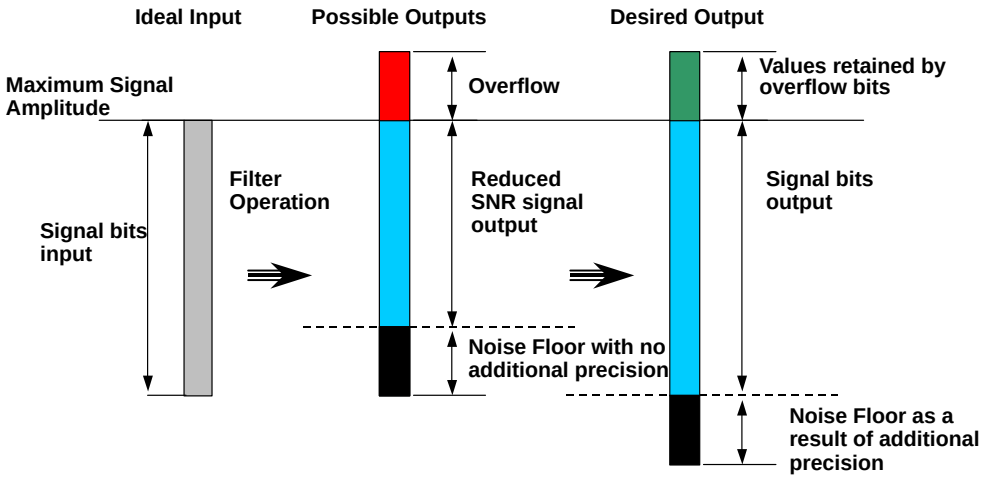


Figure 1-13. TAS5028 Digital Audio Processing

Similarly, the TAS5028 carries additional precision in the form of overflow bits to permit the value of intermediate calculations to exceed the input precision without clipping. The TAS5028 advanced digital audio processor achieves both of these important performance capabilities by using a high performance digital audio processing architecture with a 48-bit data path, 28-bit filter coefficients, and a 76-bit accumulator.

1.6 Input Crossbar Mixer

The TAS5028 has a full 8x8 input crossbar mixer. This mixer permits each signal processing channel input to be any ratio of any of the eight input channels. The control parameters for the input crossbar mixer are programmable via the I^2C interface. See the *Input Mixer Register (0x41-0x48, channels 1-8)* section.

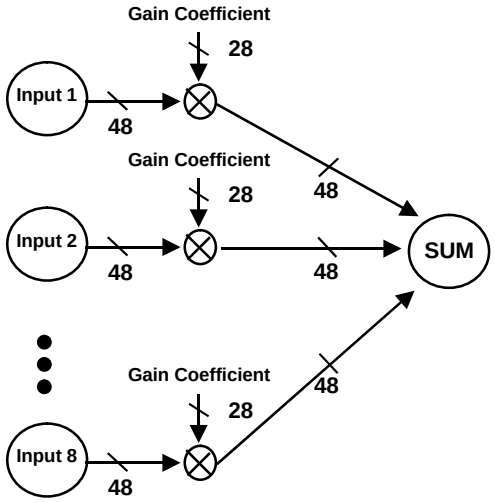


Figure 1-14. Input Crossbar Mixer

1.7 Bass and Treble Controls

From 32-kHz to 96-kHz data, the TAS5028 has four Bass and Treble tone controls. Each control has a ±18-dB control range with selectable corner frequencies and 2nd order slopes. These controls operate four channel groups:

- L, R & C (Channels 1, 2, and 7)
- LS, RS (Channels 3 and 4)
- LBS, RBS (or alternately called L and R Lineout.) (Channels 5 and 6)
- Sub (Channel 8)

For 176.4 kHz and 192 kHz data, the TAS5028 has two Bass and Treble tone controls. Each control has a ±18-dB I^2C control range with selectable corner frequencies and 2nd order slopes. These controls operate two channel groups:

- L & R
- Sub

The bass and treble filters utilize a soft update rate that does not produce artifacts during adjustment.

Table 1-3. Bass and Treble Filter Selections

FS (kHz)	3-dB CORNER FREQUENCIES									
	FILTER SET 1	FILTER SET 1	FILTER SET 2	FILTER SET 2	FILTER SET 3	FILTER SET 3	FILTER SET 4	FILTER SET 4	FILTER SET 5	FILTER SET 5
	BASS	TREBLE	BASS	TREBLE	BASS	TREBLE	BASS	TREBLE	BASS	TREBLE
32	42	917	83	1833	125	3000	146	3667	167	4333
38	49	1088	99	2177	148	3562	173	4354	198	5146
44.1	57	1263	115	2527	172	4134	201	5053	230	5972
48	63	1375	125	2750	188	4500	219	5500	250	6500
88.2	115	2527	230	5053	345	8269	402	10106	459	11944
96	125	2750	250	5500	375	9000	438	11000	500	13000
176.4	230	5053	459	10106	689	16538	804	20213	919	23888
192	250	5500	500	11000	750	18000	875	22000	1000	26000

The I^2C registers that control Bass and Treble are:

- Bass and Treble By-Pass Register (0x89 – 0x90, channels 1-8)
- Bass and Treble Slew Rates (0xD0)
- Bass Filter Sets 1-5 (0xDA)
- Bass Filter Index (0xDB)
- Treble Filter Sets 1-5 (0xDC)
- Treble Filter Index (0xDD)

1.8 Volume, Auto Mute, and Mute

The TAS5028 provides individual channel and master volume controls. Each control provides an adjustment range of +18.0618 dB to –100 dB in 0.25 dB increments. This permits a total volume device control range of +36 dB to –100 dB plus mute. The master volume control can be configured to control six or eight channels.

The TAS5028 has a master soft mute control that can be enabled by a terminal or I^2C command. The device also has individual channel soft mute controls that can are enabled via I^2C.

The soft volume and mute update rates are programmable. The soft adjustments are performed using a soft gain linear update with an I^2C programmable linear step size at a fixed temporal rate. The linear soft gain step size can be varied from 0.5 to 0.003906.

Table 1-4. Linear Gain Step Size

STEP SIZE (GAIN)	0.5	0.25	0.125	0.0625	0.03125	0.015625	0.007813	0.003906
Time to go from 36.124 db to -127 dB in ms	10.67	21.33	42.67	85.34	170.67	341.35	682.70	1365.4
Time to go from 18.062 db to -127 dB in ms	1.33	2.67	5.33	10.67	21.33	42.67	85.33	170.67
Time to go from 0 db to -127 dB in ms	0.17	0.33	0.67	1.33	2.67	5.33	10.67	21.33

1.8.1 Auto Mute and Mute

The TAS5028 has individual channel automute controls that are enabled via the I^2C interface. There are two separate detectors used to trigger the automute:

- Input Auto Mute: All channels are muted when all 8 inputs to the TAS5028 are less in magnitude than the input threshold value for a programmable amount of time.
- Output Auto Mute: A single channel is muted when the output of the DAP section is less in magnitude than the input threshold value for a programmable amount of time.

The detection period and thresholds for these two detectors are the same.

This time interval is selectable via I^2C to be from 1 ms. to 110 ms. The increments of time are 1, 2, 3, 4, 5, 10, 20, 30, 40, 50, 60, 70, 80, 90, 100, and 110 ms. This interval is independent of the sample rate. The default value is mask programmable.

The input threshold value is an unsigned magnitude that is expressed as a bit position. This value is adjustable via I^2C. The range of the input threshold adjustment is from below the LSB (bit position 0) to below bit position 12 in a 24 bit input data word (bit positions 8 to 20 in the DSPE). This provides an input threshold that can be adjusted for 12 to 24 bits of data. The default value is mask programmable.

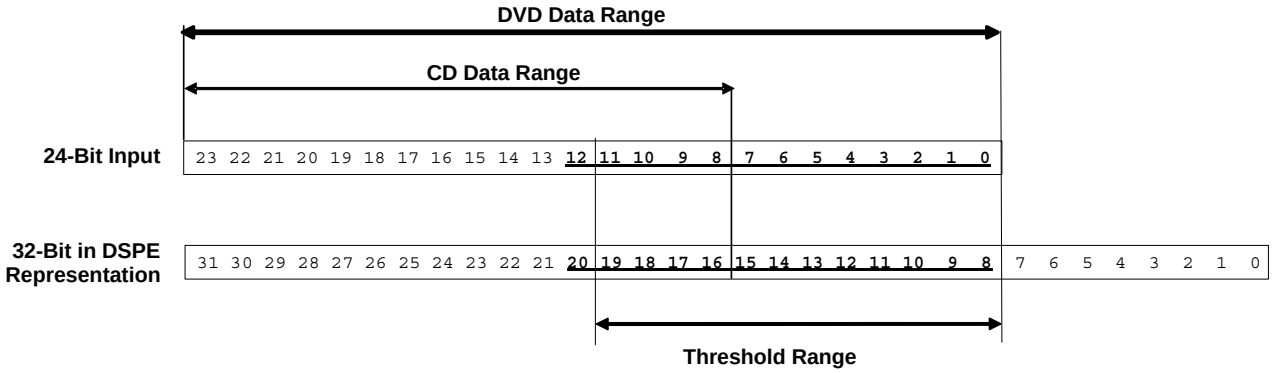


Figure 1-15. Auto Mute Threshold

The auto mute state is exited when the TAS5028 receives one sample that is greater that the output threshold.

The output threshold can be one of two values:

- Equal to the input threshold
- 6 dB (one bit position) greater than the input threshold

The value for the output threshold is selectable via I^2C. The default value is mask programmable.

The system latency enables the data value that is above the threshold to be preserved and output.

A mute command initiated by automute, master mute, individual I^2C mute, the AM interference mute sequence, or the bank switch mute sequence overrides an unmute command or a volume command. While a mute command is activated, the commanded channels transition to the mute state. When a channel is unmuted, it goes to the last commanded volume setting that has been received for that channel.

1.9 Output Mixer

The TAS5028 provides an 8x2 output mixer for channels 1, 2, 3, 4, 5, and 6. For channels 7 and 8 the TAS5028 provides an 8x3 output mixer. These mixers allow each output to be any ratio of any two (three) signal processed channels. The control parameters for the output crossbar mixer are programmable via the I^2C interface.

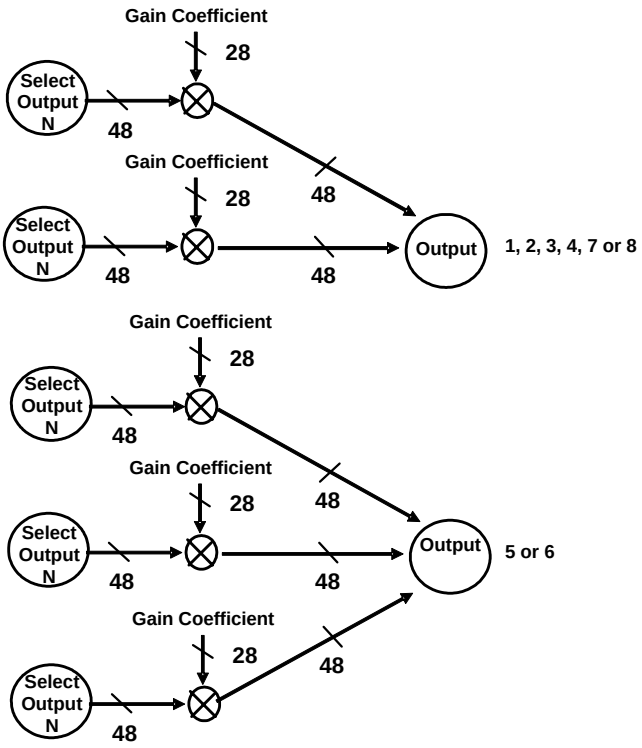


Figure 1-16. Output Mixers

1.10 PWM

The TAS5028 has eight channels of high performance digital PWM Modulators that are designed to drive switching output stages (backends) in both single-ended (SE) and H-bridge (bridge tied load) configuration. The TAS5028 device uses noise-shaping and sophisticated error correction algorithms to achieve high power efficiency and high-performance digital audio reproduction. The TAS5028 uses an AD1 PWM modulation combined with a 5^{th} order noise shaper to provide a 102-dB SNR from 20 to 20 kHz.

The PWM section accepts 32-bit PCM data from the DAP and outputs eight PWM audio output channels configurable as either:

- Six channels to drive power stages + two channels to drive a differential input active filter to provide a separately controllable stereo line out
- Eight channels to drive power stages

The TAS5028 PWM section output supports both single-ended and bridge-tied loads.

The PWM section provides a headphone PWM output to drive an external differential amplifier like the TPA112. The headphone circuit uses the PWM modulator for channels 1 and 2. The headphone will not operate while the six or eight backend drive channels are operating. The headphone will be enabled via a headphone select terminal or I^2C command.

The PWM section has individual channel dc blocking filters that can be enabled and disabled. The filter cutoff frequency is less than 1 Hz.

The PWM section has individual channel de-emphasis filters for 32, 44.1, and 48 kHz that can be enabled and disabled.

The interpolator, noise shaper, and PWM sections provide a PWM output with the following features:

- Up to 8x over sampling.
  - 8x at F_S = 32 kHz, 38 kHz, 44.1 kHz, 48 kHz
  - 4x at F_S = 88.2 kHz, 96 kHz
  - 2x at F_S = 176.4 kHz, 192 kHz
- 5th order noise shaping
- 100-dB dynamic range 0 – 20 kHz (TAS5028 + TAS5111 system measured at speaker terminals)
- THD < 0.01%
- Adjustable maximum modulation limit of 93.8% to 99.2%
- 3.3-V digital signal

1.10.1 DC Blocking (High-Pass Enable / Disable)

Each input channel incorporates a first order digital high-pass filter to block potential dc components. The filter –3 dB point is approximately 0.89-Hz at 44.1-kHz sampling rate. The high-pass filter can be enabled and disabled via the I^2C interface.

1.10.2 De-Emphasis Filter

For audio sources that have been pre-emphasized, a precision 50 μs/15 μs de-emphasis filter is provided to support the sampling rates of 32 kHz, 44.1 kHz, and 48 kHz. Figure 1-17 shows a graph of the de-emphasis filtering characteristics. De-emphasis is set using two bits in the system control register.

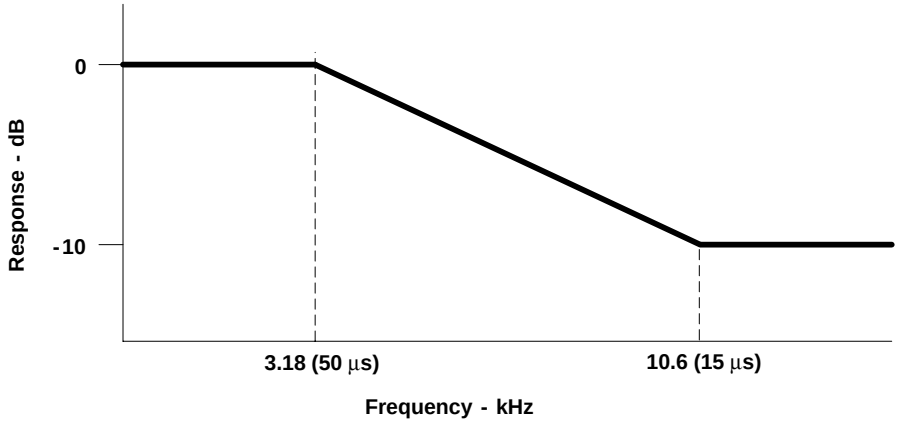


Figure 1-17. De-emphasis Filter Characteristics

1.10.3 AM Interference Avoidance

Digital amplifiers can degrade AM reception as a result of their RF emissions. Texas Instruments patented AM interference avoidance circuit provides a flexible system solution for a wide variety of digital audio architectures. During AM reception, the TAS5028 adjusts the radiated emissions to provide an emission clear zone for the tuned AM frequency. The inputs to the TAS5028 for this operation are the tuned AM frequency, the IF frequency, and the sample rate. The sample rate is automatically detected.

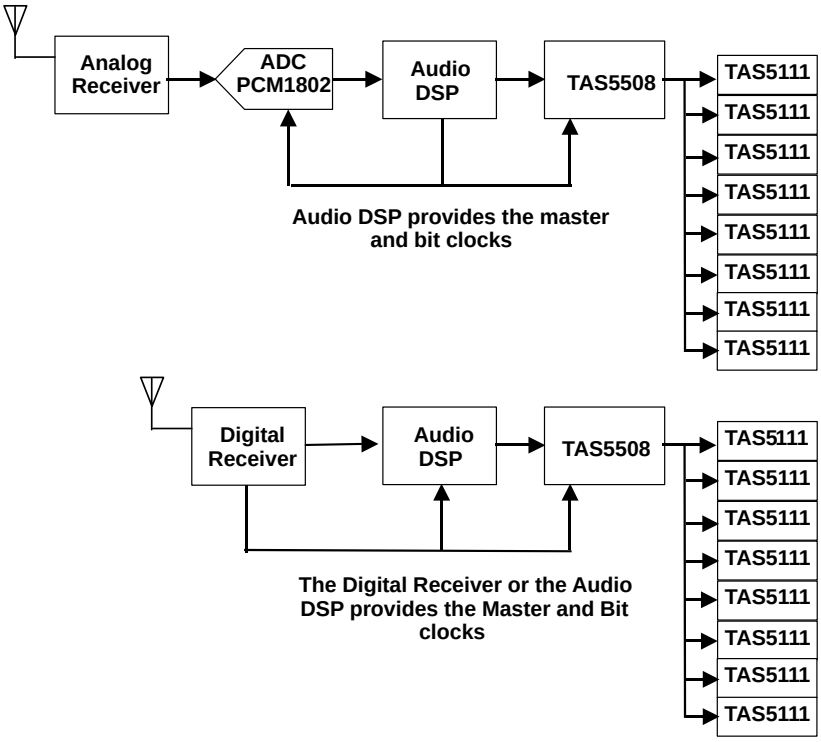


Figure 1-18. Block Diagrams of Typical Systems Requiring TAS5028 Automatic AM Interference Avoidance Circuit

2 TAS5028 Controls and Status

The TAS5028 provides control and status information from both the I^2C registers and device pins.

This section describes some of these controls and status functions. The I^2C summary and detailed register descriptions are contained in sections at the end of this document.

2.1 I^2C Status Registers

The TAS5028 has two status registers that provide general device information. These are the General Status Register 0 (0x01) and the Error Status Register (0x02).

2.1.1 General Status Register (0x01)

- Device identification code
- Clip indicator – The TAS5028 has a clipping indicator. Writing to the register clears the indicator.
- Bank switching is busy

2.1.2 Error Status Register (0x02)

- No internal errors (the valid signal is high)
- A clock error has occurred – These are sticky bits that are cleared by writing to the register.
  - LRCLK error – When the number of MCLKs per LRCLK is incorrect
  - SCLK error – When the number of SCLKS per LRCLK is incorrect
  - Frame slip – When the number of MCLKs per LRCLK changes by more than 10 MCLK cycles
  - PLL phase-lock error
- This error status register is normally used for system development only.

2.2 TAS5028 Pin Controls

The TAS5028 provide a number of terminal controls to manage the device operation. These controls are:

- $\overline{\text{RESET}}$
- $\overline{\text{PDN}}$
- $\overline{\text{BKND_ERR}}$
- $\overline{\text{HP_SEL}}$
- $\overline{\text{MUTE}}$

2.2.1 Reset ($\overline{\text{RESET}}$)

The TAS5028 is placed in the reset mode by setting the $\overline{\text{RESET}}$ terminal low or by the power up reset circuitry when power is applied.

$\overline{\text{RESET}}$ is an asynchronous control signal that restores the TAS5028 to the hard mute state (M). Master volume is immediately set to full attenuation (there is no ramp down). Reset initiates the device reset without an MCLK input. As long as the $\overline{\text{RESET}}$ terminal is held low, the device is in the reset state. During reset, all I^2C and serial data bus operations are ignored.

Table 2-1 shows the device output signals while $\overline{\text{RESET}}$ is active.

Table 2-1. Device Outputs During Reset

SIGNAL	SIGNAL STATE
Valid	Low
PWM P-outputs	Low (M-State)
PWM M-outputs	Low (M-State)
SDA	Signal Input (not driven)

Because $\overline{\text{RESET}}$ is an asynchronous signal, clicks and pops produced during the application (the leading edge) of $\overline{\text{RESET}}$ cannot be avoided. However, the transition from the hard mute state (M) to the operational state is performed using a quiet start up sequence to minimize noise. This control uses the PWM reset and unmute sequence to shut down and start up the PWM. A detailed description of these sequences is contained in the PWM section. If a completely quiet reset or power down sequence is desired, $\overline{\text{MUTE}}$ should be applied before applying $\overline{\text{RESET}}$.

The rising edge of the reset pulse begins device initialization before the transition to the operational mode. During device initialization, all controls are reset to their initial states. Table 2-2 shows the default control settings following a reset.

Table 2-2. Values Set During Reset

CONTROL	SETTING
Clock register	Not valid
High pass	Disabled
Unmute from clock error	Hard unmute
Post DAP detection automute	Enabled
Eight Ch PreDAP detection automute	Enabled
De-emphasis	De-emphasis disabled
Channel configuration control	Configured for the default setting
Headphone configuration control	Configured for the default setting
Serial data interface format	I^2S 24 bit
Individual channel mute	No channels are muted
Automute delay	5 ms
Automute threshold 1	< 8 bits
Automute threshold 2	Same as automute threshold 1
Modulation limit	Maximum modulation limit of 97.7%
Six (or eight – low) channel configuration	Eight channels
Slew rate limit	Disengaged for all channels
Interchannel delay	-32, 0, –16, 16, –24, 8, –8, -24
Shutdown PWM on error	Enabled
Volume and mute update rate	Volume ramp 85 ms
Treble and bass slew rate	Update every 1.31 ms
Bank switching	Manual bank selection is enabled
Auto bank switching map	All channels use Bank 1
Input mixer coefficients	Input N -> Channel N, no attenuation
Output mixer coefficients	Channel N -> Output N, no attenuation
Subwoofer sum into Ch1 and 2 (5508)	Gain of 0
Ch1 and 2 sum in subwoofer (5508)	Gain of 0
Bass and treble bypass	Gain of 1
Bass and treble Inline	Gain of 0
Master volume	Mute
Individual channel volumes	0 dB
All bass and treble Indexes	0x12 neutral
Treble filter sets	Filter Set 3
Bass filter sets	Filter Set 3
AM interference enable	Disabled
AM interference IF	455
AM interference select sequence	1
Tuned freq and mode	0000 , BCD

After the initialization time, the TAS5028 starts the transition to the operational state with the Master volume set at mute.

Since the TAS5028 has an external crystal time base, following the release of RESET, the TAS5028 sets the MCLK and data rates and perform the initialization sequences. The PWM outputs are held at a mute state until the master volume is set to a value other than mute via I^2C.

2.2.2 Power Down ($\overline{PDN}$)

TheTAS5028 can be placed into the power down mode by holding the $\overline{PDN}$ terminal low. When power down mode is entered, both the PLL and the oscillator are shut down. Volume is immediately set to full attenuation (there is no ramp down). This control uses the PWM mute sequence that provides a low click and pop transition to the hard mute state (M). A detailed description of the PWM mute sequence is contained in the PWM section.

Power down is an asynchronous operation that does not require MCLK to go into the power down state. To initiate the power-up sequence requires MCLK to be operational and the TAS5028 to receive 5 MCLKs prior to the release of $\overline{PDN}$.

As long as the $\overline{PDN}$ terminal is held low the device is in the power down state with the PWM outputs in a hard mute (M) state. During power down, all I^2C and serial data bus operations are ignored. Table 2-3 shows the device output signals while $\overline{PDN}$ is active.

Table 2-3. Device Outputs During Power Down

SIGNAL	SIGNAL STATE
Valid	Low
PWM P-outputs	M-state = low
PWM M-outputs	M-state = low
SDA	Signal input

Following the application of $\overline{PDN}$, the TAS5028 does not perform a quiet shutdown to prevent clicks and pops produced during the application (the leading edge) of this command. The application of $\overline{PDN}$ immediately performs a PWM stop. A quiet stop sequence can be performed by first applying $\overline{MUTE}$ before $\overline{PDN}$.

When $\overline{PDN}$ is released, the system goes to the end state specified by $\overline{MUTE}$ and $\overline{BKND_ERR}$ pins and the I^2C register settings.

The crystal time base allows the TAS5028 to determine the CLK rates. Once these rates are determined, the TAS5028 unmutes the audio.

2.2.3 Backend Error ($\overline{BKND_ERR}$)

Backend error is used to provide error management for backend error conditions. Backend error is a level sensitive signal. Backend error can be initiated by bringing the $\overline{BKND_ERR}$ terminal low for a minimum 5 MCLK cycles. When $\overline{BKND_ERR}$ is brought low, the PWM sets either six or eight channels into the PWM backend error state. This state is described in the PWM section. Once the backend error sequence is initiated, a delay of 5 ms is performed before the system starts the output re-initialization sequence. After the initialization time, the TAS5028 begins normal operation. Backend error does not affect other PWM modulator operations

The number of channels that are affected by the $\overline{BKND_ERR}$ signal is dependent upon the 6-channel configuration signal. If the I^2C setting 6-channel configuration is false, the TAS5028 places all eight PWM outputs in the PWM backend error state, while not affecting any other internal settings or operations. If the I^2C setting six configuration is true, the TAS5028 brings the PWM outputs 1-6 to a backend error state, while not affecting any other internal settings or operations. Table 2-4 shows the device output signal states during backend error.

Table 2-4. Device Outputs During Backend Error

SIGNAL	SIGNAL STATE
Valid	Low
PWM P-outputs	M-State - low
PWM M-outputs	M-State - low
HPPWM P-outputs	M-State - low
HPPWM M-outputs	M-State - low
SDA	Signal Input (not driven)

2.2.4 Speaker / Headphone Selector ($\overline{HP_SEL}$)

The $\overline{HP_SEL}$ terminal enables the headphone output or the speaker outputs. The headphone output receives the processed data output from DAP and PWM channels 1 and 2.

In 6-channel configuration this feature does not affect the two lineout channels.

When low, the headphone output is enabled. In this mode the speaker outputs are disabled. When high, the speaker outputs are enabled and the headphone is disabled.

Changes in the pin logic level results in a state change sequence using soft mute to the hard mute (M) state for both speaker and headphone followed by a soft unmute.

When $\overline{HP_SEL}$ is low, the configuration of channels 1 and 2 are defined by the headphone configuration register. When HP_SEL is high, the channel 1 and 2 configuration registers define the configuration of channels 1 and 2.

2.2.5 Mute ($\overline{MUTE}$)

The mute control provides a noiseless volume ramp to silence. Releasing mute provides a noiseless ramp to previous volume. The TAS55508 has both a master and individual channel mute commands. A terminal is also provided for the master MUTE. The low active master Mute I^2C register and the $\overline{MUTE}$ terminal are logically Or'ed together. If either is set to low, a mute on all channels is performed. The master mute command operates on all channels regardless on whether the system is in six or eight channel configuration.

When MUTE is invoked, the PWM output stops switching and then goes to an idle state.

The master Mute terminal is used to support a variety of other operations in the TAS5028, such as setting the inter-channel delay, the serial interface format, and the clock rates. A mute command by the master mute terminal, individual I^2C mute, the AM interference mute sequence, the bank switch mute sequence, or automute overrides an unmute command or a volume command. While a mute is active, the commanded channels will be placed in a mute state. When a channel is unmuted, it goes to the last commanded volume setting that has been received for that channel.

2.3 Device Configuration Controls

The TAS5028 provides a number of system configuration controls that are set at initialization and following a reset.

- Channel Configuration
- Headphone Configuration
- Audio System Configurations
- Recovery from Clock Error
- Volume and Mute Update Rate
- Modulation Index Limit
- Inter-channel Delay
- Master Clock and Data Rate Controls
- Bank Controls

2.3.1 Channel Configuration Registers

In order for the TAS5028 to have full control of the power stages, registers 0x05 to 0x0C must be programmed to reflect the proper power stage and how each one should be controlled. Channel configuration registers consist of eight registers, one for each channel.

The primary reason for using these registers is that different power stages require different handling during start up, mute/unmute, shutdown, and error recovery. The TAS5028 must select the sequence that gives the best click and pop performance and insure that the bootstrap capacitor is charged correctly during start up. This sequence depends on which power stage is present at the TAS5028 output.

Table 2-5. Description of the Channel Configuration Registers (0x05 to 0x0C)

BIT	DESCRIPTION
D7	Enable/disable error recovery sequence. In case the $\overline{\text{BKND_RECOVERY}}$ pin is pulled low, this register determines if this channel is to follow the error recovery sequence or to continue with no interruption.
D6	Determines if the power stage needs the TAS5028 VALID pin to go low to reset the power stage. Some power stages can be reset by a combination of PWM signals. For these devices, it is recommended to set this bit low, since the VALID pin is shared for power stages. This provides better control of each power stage.
D5	Determines if the power stage needs the TAS5028 VALID pin to go low to mute the power stage. Some power stages can be muted by a combination of PWM signals. For these devices, it is recommended to set this bit low, since the VALID pin is shared for power stages. This provides better control of each power stage.
D4	Inverts the PWM output. Inverting the PWM output can be an advantage if the power stage input pin are opposite the TAS5028 PWM pinout. This makes routing on the PCB easier. To keep the phase of the output the speaker terminals must also be inverted.
D3	The power stage TAS5182 has a special PWM input. To ensure that the TAS5028 has full control in all occasions, the PWM output must be remapped.
D2	Can be used to handle click and pop for some applications.
D1	This bit is normally used together with D2. For some power stages, both PWM signals must be high to get the desired operation of both speaker outputs to be low. This bit sets the PWM outputs high-high during mute.
D0	Not used

Table 2-6 lists the optimal setting for each output stage configuration. Note that the default value is applicable in all configurations except the TAS5182 SE/BTL configuration.

Table 2-6. Recommended TAS5028 Configurations for Texas Instruments Power Stages

DEVICE	ERROR RECOVERY	CONFIGURATION	D7	D6	D5	D4	D3	D2	D1	D0
Default	**RES**	**BTL**	**1**	**1**	**1**	**0**	**0**	**0**	**0**	**0**
TAS5111	RES	BTL	1	1	1	0	0	0	0	0
		SE	1	1	1	0	0	0	0	0
	AUT	BTL	0	1	1	0	0	0	0	0
		SE	0	1	1	0	0	0	0	0
TAS5112	RES	BTL	1	1	0	0	0	0	0	0
		SE	1	1	0	0	0	0	0	0
	AUT	BTL	0	1	0	0	0	0	0	0
		SE	0	1	0	0	0	0	0	0
TAS5182	RES	BTL	1	1	1	0	1	0	0	0
		SE	1	1	1	0	1	0	0	0

RES: The output stage requires $\overline{\text{VALID}}$ to go low to recover from a shutdown.

AUT: The power stage can auto recover from a shutdown.

BTL: Bridge tied load configuration

SE: Single-ended configuration

2.3.2 Headphone Configuration Registers

The headphone configuration controls are identical to the speaker configuration controls. The headphone configuration control settings are used in place of the speaker configuration control settings for channels 1 and 2 when the headphones are selected. In reality however, there is only one used configuration setting for headphones and that is the default setting.

2.3.3 Audio System Configurations

The TAS5028 can be configured to comply with various audio systems: 5.1-channel system, 6-channel system, 7.1-channel system and 8-channel system.

The audio system configuration is set in the General Control Register (0xE0). Bits D31 – D2 must be zero and D0 is don't care.

D1 Sets number of speakers in the system, including possible line outputs

D1 must be configured as the following according to the audio system in the application:

Table 2-7. Audio System Configuration (General Control Register 0xE0)

AUDIO SYSTEM	D31-D2	D1	D0
DEFAULT	**0**	**0**	**X**
6 channel or 5.1	0	1	X
8 channel or 7.1	0	0	X

2.3.3.1 Using Line Outputs in 6-Channel Configurations

The audio system can be configured for a 6-channel configuration (with 2 line outs) by writing a 1 to bit D1 of register 0xE0 (General Control Register). In this configuration, channel 5 and 6 processing are exactly the same as the other channels, except that Master Volume has no effect.

Note that in 6-channel configuration, channels 5 and 6 are unaffected by backend error ($\overline{\text{BKND_ERR}}$ goes low).

To use channels 5 and 6 as dry unprocessed line outs, the following setup should be done:

- Channel 5 volume and channel 6 volume should be set for a constant output such as 0 dB.
- Bass and Treble for channels 5 and 6 can be used if desired.
- If a down mix is desired on the channel 5 and 6 as line out, the down mixing can be performed using the channel 5 and channel 6 input mixers.

2.3.4 Recovery from Clock Error

The TAS5028 can be set to either perform a volume ramp up during the recovery sequence of a clock error or to simply come up in the last state (or desired state if a volume or tone update was in progress). This feature is enabled via I^2C system control register 0x03.

2.3.5 Volume and Mute Update Rate

The TAS5028 has fixed soft volume and mute ramp durations. The ramps are linear. The soft volume and mute ramp rates are adjustable by programming the I^2C register 0xD0 for the appropriate number of steps to be 512, 1024, or 2048. The update is performed at a fixed rate regardless of the sample rate.

- In normal speed, the update rate is 1 step every 4 / Fs seconds.
- In double speed, the update is 1 step every 8 / Fs seconds.
- In quad speed, the update is 1 step every 16 / Fs seconds.

Because of processor loading, the update rate can increase for some increments by +1/Fs to +3/Fs. However, the variance of the total time to go from +18 dB to mute is less than 25%.

Table 2-8. Volume Ramp Rates in ms

NUMBER OF STEPS	SAMPLE RATE (KHZ)	
	44.1, 88.2, 176.4	32, 48, 96, 192
512	46.44 ms	42.67 ms
1024	92.88 ms	85.33 ms
2048	185.76 ms	170.67 ms

2.3.6 Modulation Index Limit

PWM modulation is a linear function of the audio signal. When the audio signal is 0, the PWM modulation is 50%. When the audio signal increases towards full scale, the PWM modulation increases towards 100%. For negative signals, the PWM modulations fall below 50% towards 0%.

However, there is a limit to the maximum modulation possible. During the off-time period, the power stage connected to the TAS5028 output needs to get ready for he next on-time period. The maximum possible modulation is then set by the power stage requirements. All Texas Instruments power stages needs maximum modulation to be 97.7%. This is also the default setting of the TAS5028. Default settings can be changed in the Modulation Index Register (0x16).

Note that no change should be made to this register when using Texas Instruments power stages.

2.3.7 Inter-channel Delay

An 8-bit value can be programmed to each of the eight PWM inter-channel delay registers to add a delay per channel from 0 to 255 clock cycles. The delays correspond to cycles of the high-speed internal clock, DCLK. The default values are shown in Table 2-9.

Table 2-9. Inter-Channel Delay Default Values

I^2C SUB-ADDRESS	CHANNEL	INTER-CHANNEL DELAY DEFAULT (DCLK PERIODS)
0x1B	1	-24
0x1C	2	0
0x1D	3	-16
0x1E	4	+16
0x1F	5	-24
0x20	6	+8
0x21	7	-8
0x22	8	+24

This delay is generated in the PWM and can be changed at any time through the serial control interface I^2C registers 0x1B – 0x22. The absolute offset for channel 1 is set in I^2C sub-address 0x23.

NOTE: If used correctly, setting the PWM channel delay can optimize the performance of a pure path digital amplifier system. The setting is based upon the type of backend power device that is used and the layout. These values are set during initialization using the I^2C serial interface. Unless otherwise noted, use the default values given in Table 2-9.

2.4 Master Clock and Serial Data Rate Controls

The TAS5028 function only as a receiver of the MCLK (master clock), SCLK (shift clock), and LRCLK (left/right clock) signals that controls the flow of data on the four serial data interfaces. The 13.5-MHz external crystal allows the TAS5028 to automatically detect MCLK and the data rate.

The MCLK frequency can be 64 x Fs, 128 x Fs, 196 x Fs, 256 x Fs, 384 x Fs, 512 x Fs, or 768 x Fs.

The TAS5028 operates with the serial data interface signals LRCLK and SCLK synchronized to MCLK. However, there is no constraint as to the phase relationship of these signals. The TAS5028 accepts a 64 x Fs SCLK rate and a 1 x Fs LRCLK.

If the phase of SCLK or LRCLK drifts more than ±10 MCLK cycles since the last RESET, the TAS5028 performs a clock error and resynchronize the clock timing.

The clock and serial data interface have several control parameters:

- MCLK Ratio 64 Fs, 128 Fs, 196 Fs, 256 Fs, 384 Fs, 512 Fs, or 768 Fs) - I^2C parameter
- Data Rate 32, 38, 44.1,48, 88.2, 96, 176.4, 192 kHz - I^2C parameter
- AM Mode Enable / Disable - I^2C parameter

During AM interference avoidance, the clock control circuitry utilizes three other configuration inputs:

- Tuned AM Frequency (for AM interference avoidance) (550 - 1750 kHz) - I^2C parameter
- Frequency Set Select (1-4) - I^2C parameter
- Sample Rate - I^2C parameter or auto detected

2.4.1 PLL Operation

The TAS5028 uses two internal clocks generated by two internal phase-locked loops (PLLs), the digital PLL (DPLL) and the analog PLL (APLL). The analog PLL provides the reference clock for the PWM. The digital PLL provides the reference clock for the digital audio processor and the control logic.

The master clock MCLK input provides the input reference clock for the APLL. The external 13.5-MHz crystal provides the input reference clock for the digital PLL. The crystal provides a time base to support a number of operations, including the detection of the MCLK ratio, the data rate, and clock error conditions. The crystal time base provides a constant rate for all controls and signal timing.

Even if MCLK is not present, the TAS5028 can receive and store I^2C commands and provide status.

2.5 Bank Controls

The TAS5028 permits the user to specify and assign sample rate dependent parameters for Tone in one of three banks that can be manually selected or selected automatically based upon the data sample rate. Each bank can be enabled for one or more specific sample rates via I^2C bank control register 0x40. Each bank set holds the following values:

- Five Bass Filter-Set Selections (Register 0xDA)
- Five Treble Filter-Set Selections (Register 0xDC)

The default selection for bank control is manual bank with bank 1 selected. Note that if bank switching is used, Bank 2 and Bank 3 must be programmed on power-up since the default values are all zeroes. If bank switching is used and Bank 2 and Bank 3 are not programmed correctly, then the output of the TAS5028 could be muted when switching to those banks.

2.5.1 Manual Bank Selection

The three bank selection bits of the bank control register allow the appropriate bank to be manually selected (000 = Bank 1, 001 = Bank 2, 010 = Bank 3). In the manual mode, when a write occurs to the Bass or Treble coefficients, the current selected bank is updated. If audio data is streaming to the TAS5028, during a manual bank selection, the TAS5028 first performs a mute sequence, then performs the bank switch, and finally restores the volume using an un-mute sequence.

A mute command initiated by the bank switch mute sequence overrides an un-mute command or a volume command. While a mute is active, the commanded channels are muted. When a channel is unmated, the volume level goes to the last commanded volume setting that has been received for that channel.

If MCLK or SCLK is stopped, the TAS5028 performs a bank switch operation. If the clocks should start up once the manual bank switch command has been received, the bank switch operation is performed during the 5-ms silent start sequence.

2.5.2 Automatic Bank Selection

To enable automatic bank selection, a value of 3 is written into in the bank selection bits of the bank control register. Banks are associated with one or more sample rates by writing values into the Bank 1 or Bank 2 data rate selection registers. The automatic base selection is performed when a frequency change is detected according to the following scheme:

1. The system scans Bank 1 data rate associations to see if the Bank 1 is assigned for that data rate.
2. If Bank 1 is assigned, then the Bank 1 coefficients will be loaded.
3. If it is not then, the system scans the bank 2 to see if Bank 1 is assigned for that data rate.
4. If Bank 2 is assigned, then the Bank 2 coefficients will be loaded.
5. If it is not then, the system loads the Bank 3 coefficients.

The default is that all frequencies are enabled for Bank 1. This default is expressed as a value of all 1s in the Bank 1 auto-selection byte and all 0s in the bank 2 auto-section byte.

2.5.2.1 Coefficients Write Operations While Automatic Bank Switch Is Enabled

In automatic mode if a write occurs to the Tone coefficient, the bank that is written to is the current bank.

2.5.3 Bank Set

Bank set is used to provide a secure way to update the bank coefficients in both the manual and automatic switching modes without causing a bank switch to occur. Bank set mode does not alter the current bank register mapping. It simply enables any bank's coefficients to be updated while inhibiting any bank switches from taking place. In manual mode, this enables the coefficients to be set without switching banks. In automatic mode this prevents a clock error or data rate change from corrupting a bank coefficient write.

To update the coefficients of a bank, a value of 4, 5, or 6 is written into in the bank selection bits of the bank control register. This enables the Tone coefficient values of bank 1, 2, or 3 to be respectively updated.

Once the coefficients of the bank have been updated, the bank selection bits are then returned to the desired manual or automatic bank selection mode.

2.5.4 Bank Switch Timeline

After a bank switch is initiated (manual or automatic), no I^2C writes to the TAS5028 should occur before a minimum of 186 ms. This value is determined by the volume ramp rates for a particular sample rate.

2.5.5 Bank Switching Example 1

Problem: The audio unit containing a TAS5028 needs to handle different audio formats with different sample rates. Format #1 requires Fs = 32 kHz, Format #2 requires Fs = 44.1 kHz, and Format #3 requires Fs = 48 kHz. The sample-rate dependent parameters in the TAS5028 require different coefficients and data depending on the sample rate.

Strategy: Use the TAS5028 bank switching feature to allow for managing and switching three banks associated with the three sample rates, 32 kHz (Bank 1), 44.1 kHz (Bank 2), and 48 kHz (Bank 3).

One possible algorithm is to generate, load, and automatically manage bank switching for this problem:

- Generate bank-related coefficients (see above) for sample rates 32 kHz, 44.1 kHz, and 48 kHz and include the same in the micro-based TAS5028 I^2C firmware.
- On TAS5028 power up or reset, the micro runs the following TAS5028 Initialization code:
  - Update Bank 1 (Write 0x00048040 to register 0x40).
  - Write bank-related I^2C registers with appropriate values for Bank 1.
  - Write Bank 2 (Write 0x00058040 to register 0x40).
  - Load bank-related I^2C registers with appropriate values for Bank 2.
  - Write Bank 3 (Write 0x00068040 to register 0x40).
  - Load bank-related I^2C registers with appropriate values for Bank 3.
  - Select automatic bank switching (write 0x00038040 to register 0x40)
- Now when the audio media changes, the TAS5028 automatically detects the incoming sample rate and automatically switches to the appropriate bank.

In this example any sample rates other then 32 kHz and 44.1 kHz will use Bank 3. If other sample rates are used, then the banks need to be set-up differently.

2.5.6 Bank Switching Example 2

Problem: The audio system uses all of the sample rates supported by the TAS5028. How can the automatic bank switching be set up to handle this situation?

Strategy: Use the TAS5028 bank switching feature to allow for managing and switching three banks associated with sample rates as follows:

- Bank 1: Coefficients for 32 kHz, 38 kHz, 44.1 kHz, and 48 kHz
- Bank 2: Coefficients for 88.2kHz and 96 kHz
- Bank 3: Coefficients for 176.4 kHz and 192 kHz

One possible algorithm is to generate, load, and automatically manage bank switching for this problem:

- Generate bank-related coefficients for sample rates 48 kHz (Bank 1), 96 kHz (Bank 2), and 192 kHz (Bank 3) and include the same in the micro-based TAS5028 I^2C firmware.
- On TAS5028 power-up or reset, the micro runs the following TAS5028 Initialization code:
  - Update Bank 1 (Write 0x0004F00C to register 0x40).
  - Write bank-related I^2C registers with appropriate values for Bank 1.
  - Write Bank 2 (Write 0x0005F00C to register 0x40).
  - Load bank-related I^2C registers with appropriate values for Bank 2.
  - Write Bank 3 (Write 0x0006F00C to register 0x40).
  - Load bank-related I^2C registers with appropriate values for Bank 3.
  - Select automatic bank switching (Write 0x0003F00C to register 0x40)
- Now when the audio media changes, the TAS5028 automatically detects the incoming sample rate and automatically switches to the appropriate bank.

3 Electrical Specifications

3.1 Absolute Maximum Ratings†

		UNITS
Supply voltage, DVDD and DVD_PWM		-0.3 V to 3.6 V
Supply voltage, AVDD_PLL		-0.3 V to 3.6 V
Input voltage	3.3-V digital input	-0.5 V to DVDD + 0.5 V
	5 V tolerant(2) digital input	-0.5 V to 6 V
	1.8 V LVCMOS(3)	-0.5 V to VREF(1) + 0.5 V
Input clamp current, I_{IK} (V_I < 0 or V_I > 1.8 V		±20 mA
Output clamp current, I_{OK} (V_O < 0 or V_O > 1.8 V)		±20 mA
Operating free air temperature		0°C to 70°C
Storage temperature range, T_{stg}		-65°C to 150°C

† Stresses beyond those listed under “absolute maximum ratings” may cause permanent damage to the device. These are stress ratings only and functional operation of the device at these or any other conditions beyond those indicated under “recommended operating conditions” is not implied. Exposure to absolute-maximum-rated conditions for extended periods may affect device reliability.

NOTES: 1. VREF is a 1.8-V supply derived from regulators internal to the TAS5028 chip. VREF is on terminals VRA_PLL, VRD_PLL, VR_DPLL, VR_DIG, and VR_PWM. These terminals are provided to permit use of external filter capacitors, but should not be used to source power to external devices.

2. 5-V tolerant inputs are $\overline{\text{RESET}}$, $\overline{\text{PDN}}$, $\overline{\text{MUTE}}$, $\overline{\text{HP_SEL}}$, SCLK, LRCLK, MCLK, SDIN1, SDIN2, SDIN3, SDIN4, SDA, and SCL.

3. VRA_PLL, VRD_PLL, VR_DPLL, VR_DIG, VR_PWM

DISSIPATION RATING TABLE (High-k Board, 105°C Junction)

PACKAGE	$T_A \leq 25°C$ POWER RATING	DERATING FACTOR ABOVE $T_A = 25°C$	$T_A = 70°C$ POWER RATING
PAG	1869 mW	23.36 mW/°C	818 mW

3.2 Dynamic Performance (At Recommended Operating Conditions at 25°C)

PARAMETER	TEST CONDITIONS	MIN	NOM	MAX	UNITS
Dynamic range 32 kHz to 192 kHz	TAS5028 + TAS5111 A-weighted		102		dB
Total harmonic distortion	TAS5111 A at 1 W		0.1%		
	TAS5028 ouput		0.01%		
Frequency response	32-kHz to 96-kHz sample rates		±0.1		dB
	176.4, 192-kHz sample rates		±0.2		

3.3 Recommended Operating Conditions (Over 0°C to 70°C)

		MIN	NOM	MAX	UNITS
Digital supply voltage, DVDD and DVDD_PWM		3	3.3	3.6	V
Analog supply voltage, AVDD_PLL		3	3.3	3.6	V
High-level input voltage, V_{IH}	3.3 V	2			V
	5-V tolerant(4)	2			
	1.8-V LVCMOS (XTL_IN)	1.26			
Low-level input voltage, V_{IL}	3.3 V			0.8	V
	5-V tolerant(4)			0.8	
	1.8-V (XTL_IN)			0.54	
Operating ambient air temperature range, T_A		-20	25	70	°C
Operating junction temperature range, T_J		-20		105	°C

NOTE 4: 5-V tolerant inputs are SDA, SCL, $\overline{\text{RESET}}$, $\overline{\text{PDN}}$, $\overline{\text{MUTE}}$, $\overline{\text{HP_SEL}}$, SCLK, LRCLK, MCLK, SDIN1, SDIN2, SDIN3, and SDIN4.

3.4 Electrical Characteristics Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER			TEST CONDITIONS	MIN	TYP	MAX	UNITS
V_{OH}	High-level output voltage	3.3 V TTL and 5 V[6] tolerant	I_{OH} = -4 mA	2.4			V
		1.8-V LVCMOS (XTL_OUT)	I_{OH} = - 0.55 mA	1.44			
V_{OL}	Low-level output voltage	3.3-V TTL and 5 V[6] tolerant	I_{OL} = 4 mA			0.5	V
		1.8-V LVCMOS (XTL_OUT)	I_{OL} = 0.75 mA			0.5	
I_{OZ}	High-impedance output current	3.3-V TTL				±20	µA
I_{IL}	Low-level input current	3.3-V TTL	$V_I = V_{IL}$			±1	µA
		1.8-V LVCMOS (XTL_IN)	$V_I = V_{IL}$			±1	
		5 V tolerant[5]	V_I = 0 V DVDD = 3 V			±1	
I_{IH}	High-level input current	3.3-V TTL	$V_I = V_{IH}$			±1	µA
		1.8-V LVCMOS (XTL_IN)	$V_I = V_{IH}$			±1	
		5 V tolerant[5]	V_I = 5.5 V DVDD = 3 V			±20	
I_{DD}	Input supply current	Digital supply voltage, DVDD	Fs = 48 kHz		140		mA
			Fs = 96 kHz		150		
			Fs = 192kHz		155		
			Power down		8		
		Analog supply voltage, AVDD	Normal		20		mA
			Power down		2		

NOTES: 5. 5-V tolerant inputs are SDA, SCL, $\overline{\text{RESET}}$, $\overline{\text{PDN}}$, $\overline{\text{MUTE}}$, $\overline{\text{HP_SEL}}$, SCLK, LRCLK, MCLK, SDIN1, SDIN2, SDIN3, and SDIN4.
6. 5-V tolerant outputs are SCL and SDA

3.5 PWM Operation at Recommended Operating Conditions Over 0°C to 70°C

PARAMETER	TEST CONDITIONS	MODE	VALUE	UNITS
Output sample rate 1X – 8 x over sampled	32-kHz data rate ±4%	12 x sample rate	384	kHz
	44.1-, 88.2-, 176.4-kHz data rate ±4%	8, 4, and 2 x sample rate	352.8	kHz
	48, 96, 192 kHz data rate ±4%	8, 4, and 2 x sample rate	384	kHz

3.6 Switching Characteristics

3.6.1 Clock Signals Over Recommended Operating Conditions (Unless Otherwise Noted)

3.6.1.1 PLL Input Parameters and External Filter Components†

PARAMETER		TEST CONDITIONS	MIN	TYP	MAX	UNITS
f_{XTALI}	Frequency, XTAL IN	Only use 13.5-MHz crystal ≤1000 ppm		13.5		MHz
f_{MCLKI}	Frequency, MCLK (1 / t_{cyc2})		2		50	MHz
	MCLK duty cycle duty cycle		40%	50%	60%	
	MCLK minimum high time	≥2-V MCLK = 49.152 MHz, Within the min and max duty cycle constraints	5			ns
	MCLK minimum low time	≤0.8-V MCLK = 49.152 MHz, Within the min and max duty cycle constraints	5			ns
	LRCLK allowable drift before LRCLK reset				10	MCLKs
	External PLL filter cap C1	SMD 0603 Y5V		100		nF
	External PLL filter cap C2	SMD 0603 Y5V		10		nF
	External PLL filter resistor R	SMD 0603, metal film		200		Ω
	External VRA_PLL decoupling	SMD, Y5V		100		nF

See the TAS5028 *Example Application Schematic* section.

3.6.2 Serial Audio Port

3.6.2.1 Serial Audio Ports Slave Mode Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		TEST CONDITIONS	MIN	TYP	MAX	UNITS
f_{SCLKIN}	Frequency, SCLK 64 x fs	C_L = 30 pF	2.048		12.288	MHz
t_{su1}	Setup time, LRCLK to SCLK rising edge		10			ns
t_{h1}	Hold time, LRCLK from SCLK rising edge		10			ns
t_{su2}	Setup time, SDIN to SCLK rising edge		10			ns
t_{h2}	Hold time, SDIN from SCLK rising edge		10			ns
	LRCLK frequency		32	48	192	kHz
	SCLK duty cycle		40%	50%	60%	
	LRCLK duty cycle		40%	50%	60%	
	SCLK rising edges between LRCLK rising edges		64		64	SCLK edges
	LRCLK clock edge with respect to the falling edge of SCLK		-1/4		1/4	SCLK period

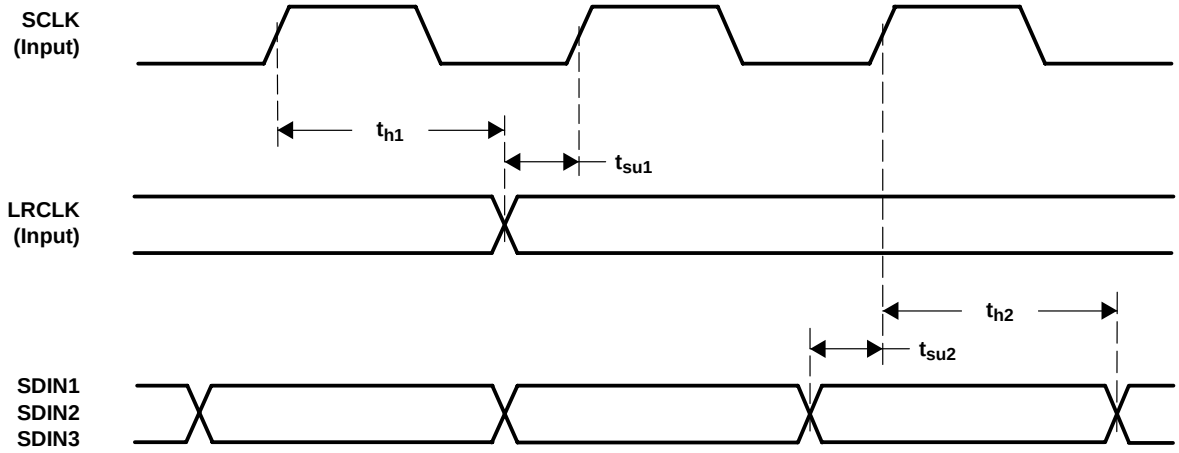


Figure 3-1. Slave Mode Serial Data Interface Timing

3.6.3 I²C Serial Control Port Operation

3.6.3.1 Timing Characteristics for I²C Interface Signals Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		TEST CONDITIONS	MIN	MAX	UNITS
f_{SCL}	Frequency, SCL	No wait states		400	kHz
$t_{w(H)}$	Pulse duration, SCL high		0.6		µs
$t_{w(L)}$	Pulse duration, SCL low		1.3		µs
t_r	Rise time, SCL and SDA			300	ns
t_f	Fall time, SCL and SDA			300	ns
t_{su1}	Setup time, SDA to SCL		100		ns
t_{h1}	Hold time, SCL to SDA		0		ns
$t_{(buf)}$	Bus free time between stop and start condition		1.3		µs
t_{su2}	Setup time, SCL to start condition		0.6		µs
t_{h2}	Hold time, start condition to SCL		0.6		µs
t_{su3}	Setup time, SCL to stop condition		0.6		µs
C_L	Load capacitance for each bus line			400	pF

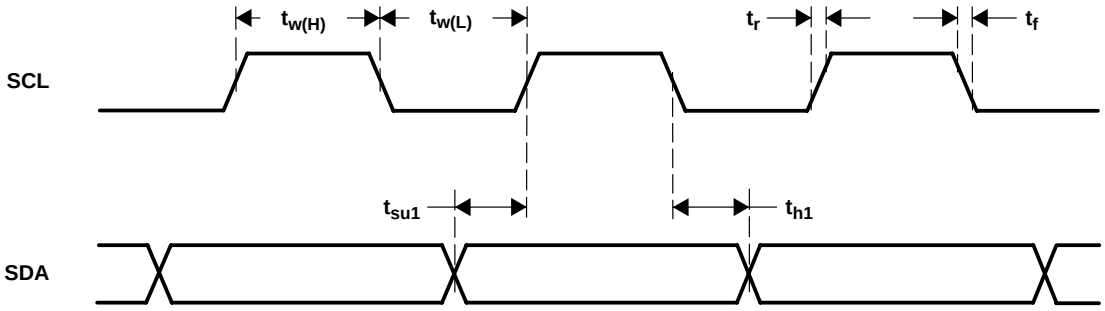


Figure 3-2. SCL and SDA Timing

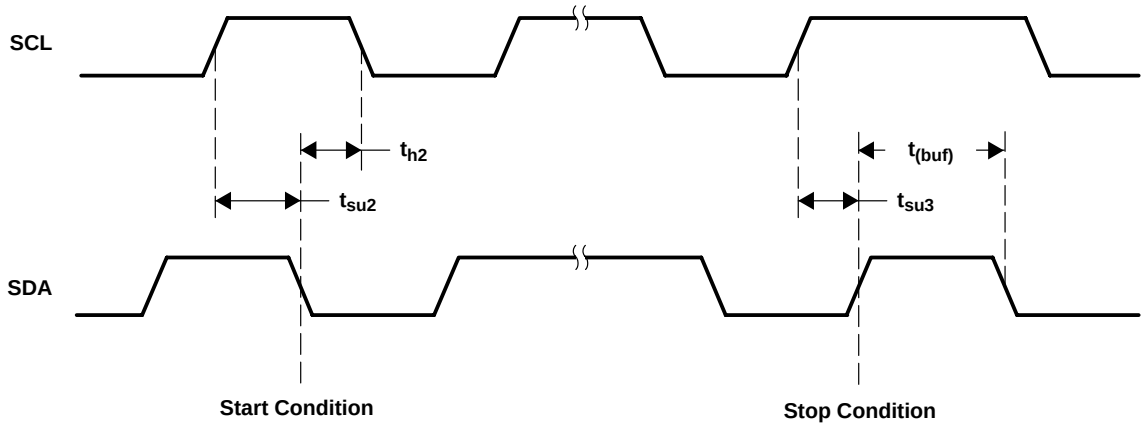


Figure 3-3. Start and Stop Conditions Timing

3.6.4 Reset Timing ($\overline{RESET}$)

3.6.4.1 Control Signal Parameters Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		MIN	TYP	MAX	UNITS
$t_{r(DMSTATE)}$	Time to $\overline{\text{M-STATE}}$ low			370	ns
$t_{w(RESET)}$	Pulse duration, RESET active	400		None	ns
$t_{r(I2C_ready)}$	Time to enable I^2C		3		ms
$t_{r(run)}$	Device startup time	10			ms

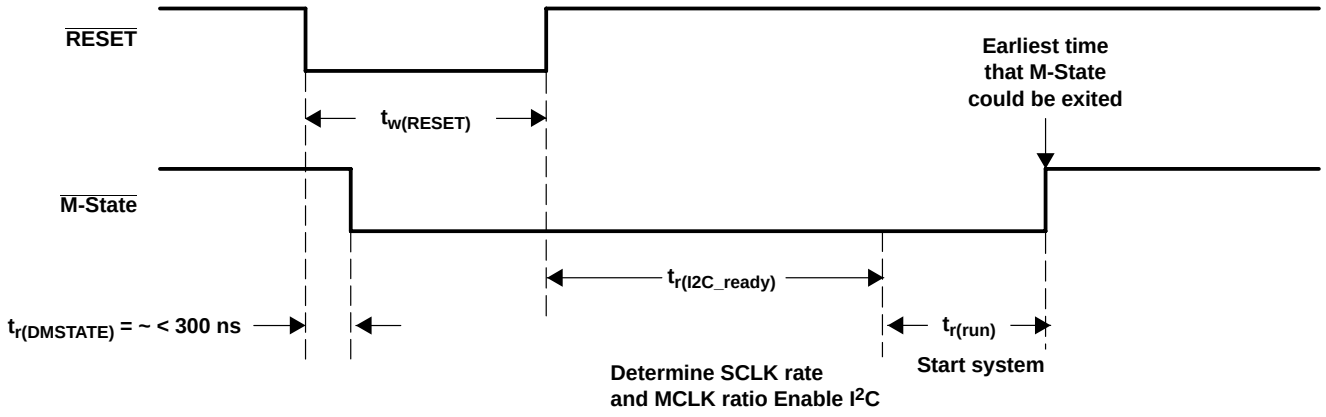


Figure 3-4. Reset Timing

Since a crystal time base is used, the system determines the CLK rates. Once the data rate and master clock ratio is determined, the system outputs audio if a master volume command is issued.

3.6.5 Power-Down ($\overline{PDN}$) Timing

3.6.5.1 Control Signal Parameters Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		MIN	TYP	MAX	UNITS
$t_{p(DMSTATE)}$	Time to $\overline{\text{M-STATE}}$ low			300	µs
	Number of MCLKs preceding the release of $\overline{\text{PDN}}$	5			
t_{su}	Device startup time		120		ms

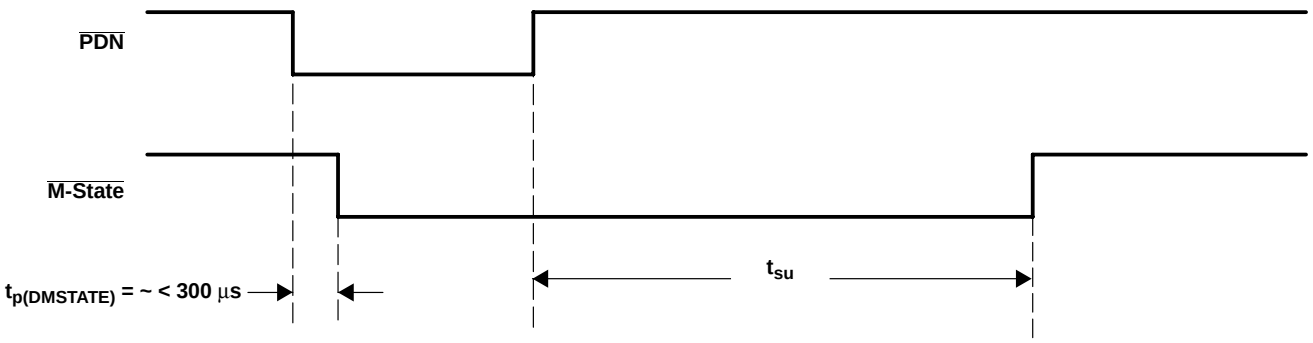


Figure 3-5. Power-Down Timing

3.6.6 Backend Error ($\overline{BKND_ERR}$)

3.6.6.1 Control Signal Parameters Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		MIN	TYP	MAX	UNITS
$t_{w(ER)}$	Pulse duration, BKND_ERR active	350		None	ns
$t_{p(valid_low)}$				<100	µs
$t_{p(valid_high)}$	I^2C programmable to be between 1 to 10 ms	-25		25	% of interval

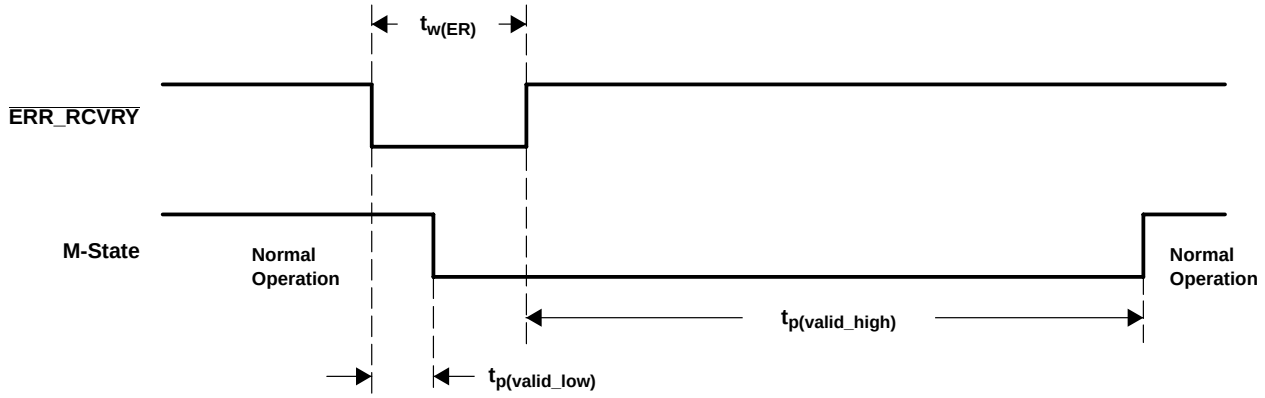


Figure 3-6. Error Recovery Timing

3.6.7 MUTE Timing—$\overline{MUTE}$

3.6.7.1 Control Signal Parameters Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		MIN	TYP	MAX	UNITS
$t_{d(VOL}$	Volume ramp time	Defined by rate setting(1)			ms

NOTE 1: See the *Volume Treble and Base Slew Rate Register (0xD0)* section.

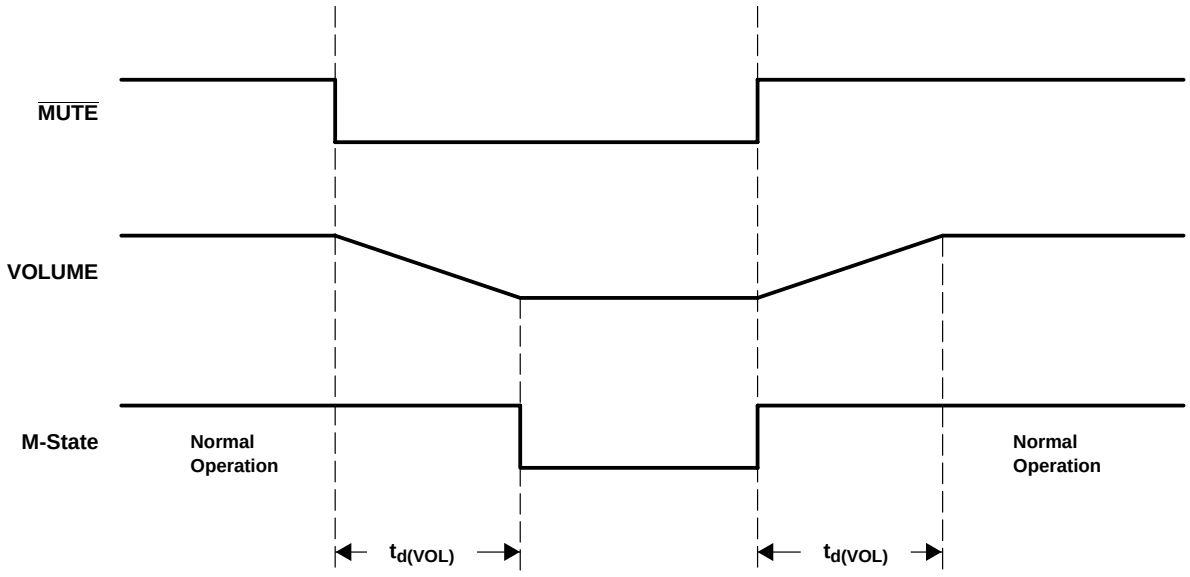


Figure 3-7. Mute Timing

3.6.8 *Headphone Select* (HP_SEL)

3.6.8.1 Control Signal Parameters Over Recommended Operating Conditions (Unless Otherwise Noted)

PARAMETER		MIN	MAX	UNITS
$t_{w(MUTE)}$	Pulse duration, HP_SEL active	350	None	ns
$t_{d(VOL)}$	Soft volume update time	Defined by rate setting(2)		ms
$t_{(SW)}$	Switch-over time	0.2	1 ms	ms

NOTE 2: See the *Volume Treble and Base Slew Rate Register (0xD0)* section.

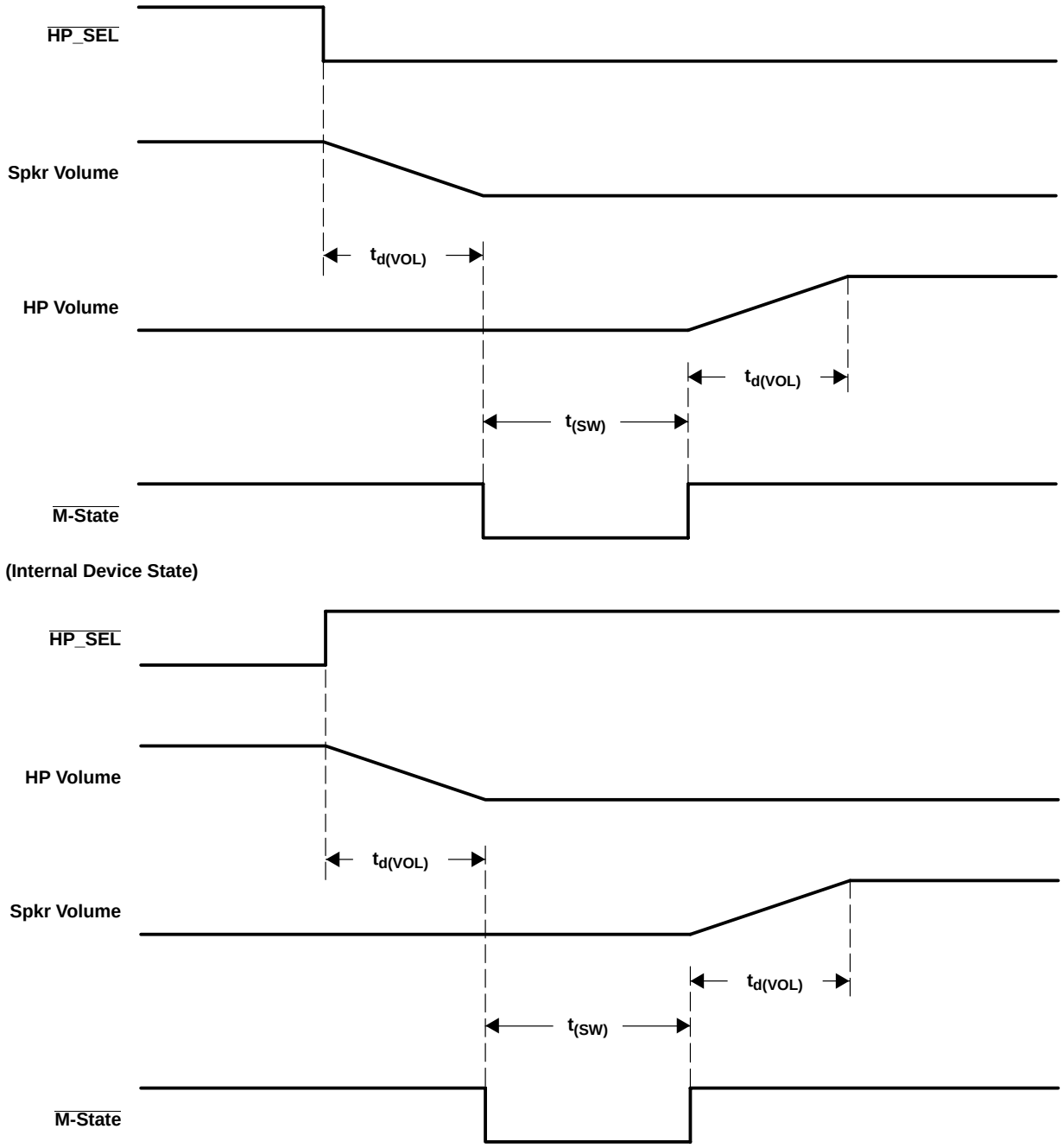


Figure 3-8. HP_SEL Timing

3.6.9 Volume Control

3.6.9.1 Control Signal Parameters Over Recommended Operating Conditions (Unless Otherwise Noted)

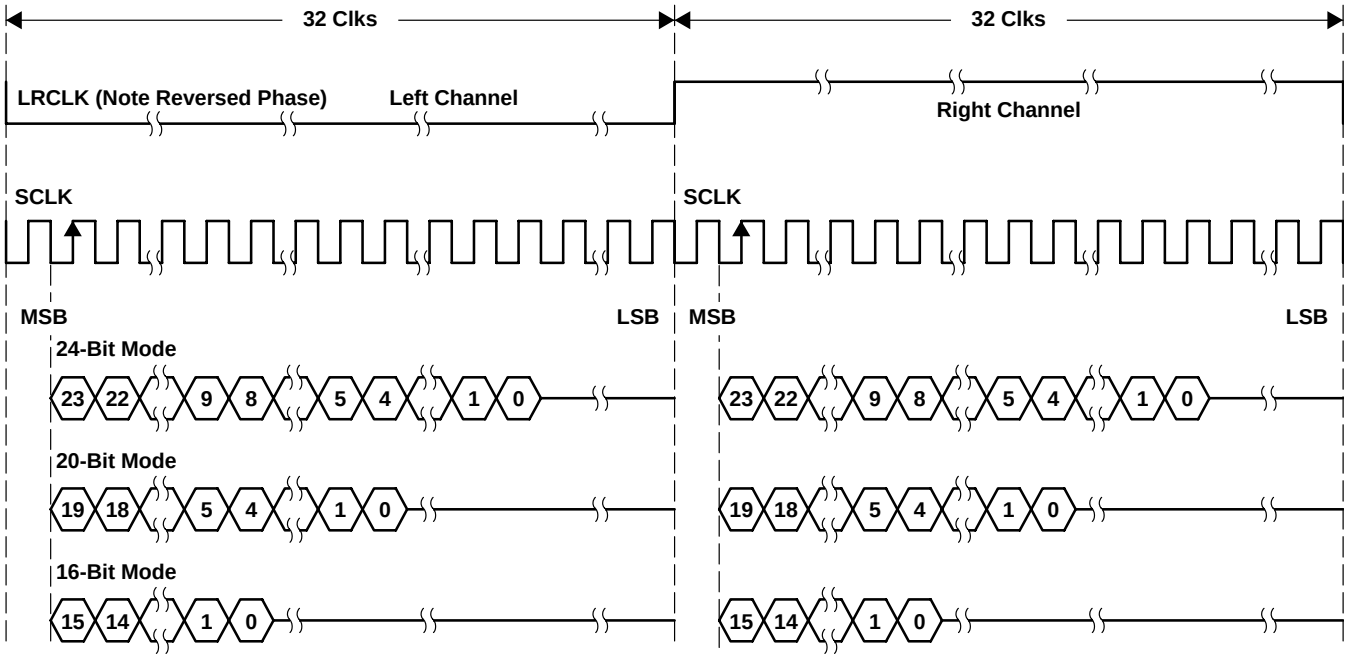
PARAMETER	TEST CONDITIONS	MIN	MAX	UNITS
Maximum attenuation before mute	Individual volume, master volume or a combination of both		-127	dB
Maximum gain	Individual volume, master volume		18	dB
Maximum volume before the onset of clipping	0-dB input, any modulation limit		0	dB

3.7 Serial Audio Interface Control and Timing

3.7.1 I^2S Timing

I^2S timing uses an LRCLK to define when the data being transmitted is for the left channel and when it is for the right channel. The LRCLK is low for the left channel and high for the right channel. A bit clock running at 64 × Fs is used to clock in the data. There is a delay of one bit clock from the time the LRCLK signal changes state to the first bit of data on the data lines. The data is written MSB first and is valid on the rising edge of bit clock. The TAS5028 masks unused trailing data bit positions.

2-Channel I^2S (Philips Format) Stereo Input



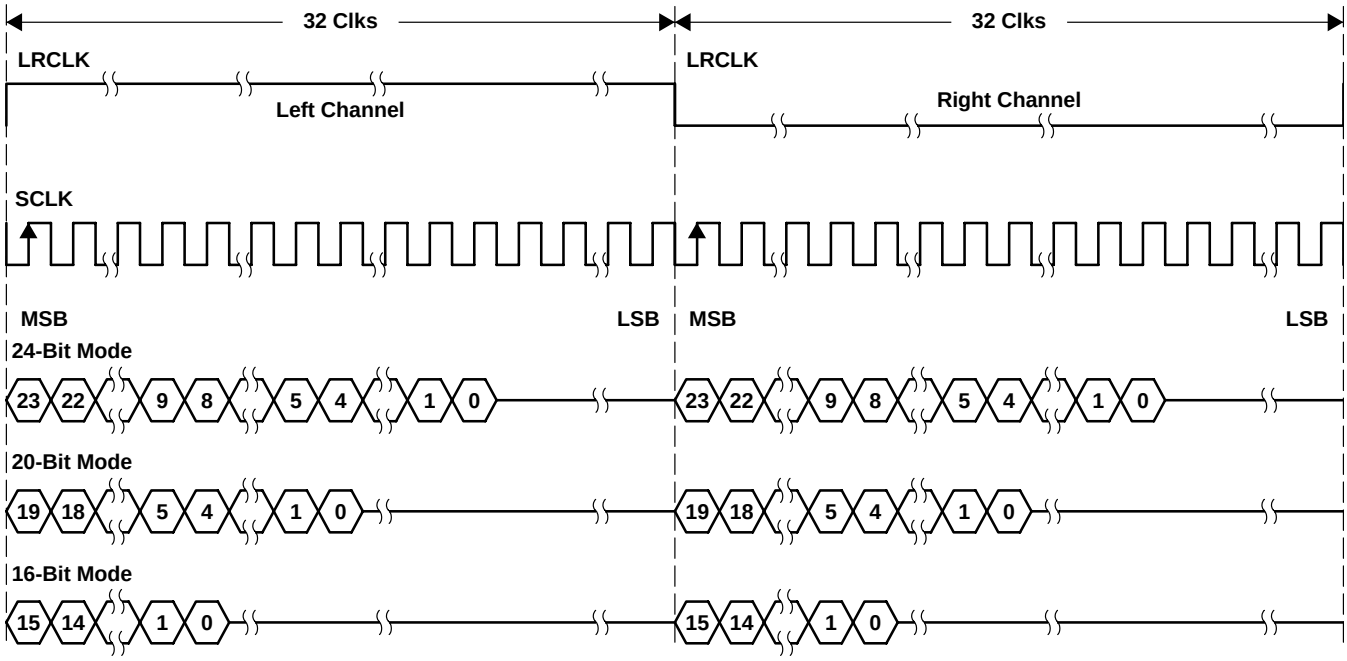
NOTE: All data presented in 2s complement form with MSB first.

Figure 3-9. I^2S Format 64 Fs Format

3.7.2 Left Justified

Left justified (LJ) timing uses an LRCLK to define when the data being transmitted is for the left channel and when it is for the right channel. The LRCLK is high for the left channel and low for the right channel. A bit clock running at 64 × Fs is used to clock in the data. The first bit of data appears on the data lines at the same time the LRCLK toggles. The data is written MSB first and is valid on the rising edge of bit clock. The TAS5028 masks unused trailing data bit positions.

2-Channel Left-Justified Stereo Input



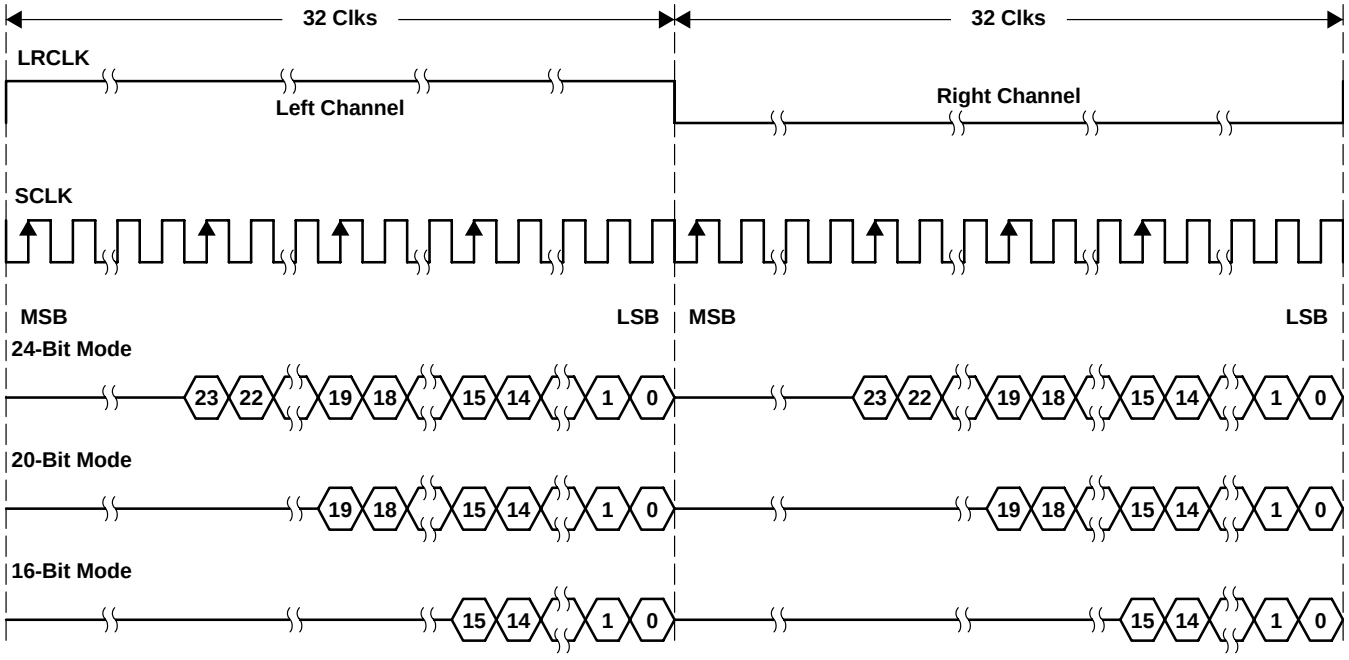
NOTE: All data presented in 2s complement form with MSB first.

Figure 3-10. Left Justified 64 Fs Format

3.7.3 Right Justified

Right justified (RJ) timing uses an LRCLK to define when the data being transmitted is for the left channel and when it is for the right channel. The LRCLK is high for the left channel and low for the right channel. A bit clock running at 64 × Fs is used to clock in the data. The first bit of data appears on the data 8-bit clock periods (for 24-bit data) after LRCLK toggles. In RJ mode the LSB of data is always clocked by the last bit clock before L/RCLK transitions. The data is written MSB first and is valid on the rising edge of bit clock. The TAS5028 masks unused leading data bit positions.

2-Channel Right-Justified (Sony Format) Stereo Input



NOTE: All data presented in 2s complement form with MSB first.

Figure 3-11. Right Justified 64 Fs Format

4 I²C Serial Control Interface (Slave Address 0x36)

The TAS5028 has a bidirectional I²C interface that compatible with the I²C (Inter IC) bus protocol and supports both 100 Kbps and 400 Kbps data transfer rates for single and multiple byte write and read operations. This is a slave only device that does not support a multi-master bus environment or wait state insertion. The control interface is used to program the registers of the device and to read device status.

The TAS5028 supports the standard-mode I²C bus operation (100 kHz maximum) and the fast I²C bus operation (400 kHz maximum). The TAS5028 performs all I²C operations without I²C wait cycles.

4.1 General I²C Operation

The I²C bus employs two signals; SDA (data) and SCL (clock), to communicate between integrated circuits in a system. Data is transferred on the bus serially one bit at a time. The address and data can be transferred in byte (8 bit) format, with the most significant bit (MSB) transferred first. In addition, each byte transferred on the bus is acknowledged by the receiving device with an acknowledge bit. Each transfer operation begins with the master device driving a start condition on the bus and ends with the master device driving a stop condition on the bus. The bus uses transitions on the data terminal (SDA) while the clock is high to indicate a start and stop conditions. A high-to-low transition on SDA indicates a start and a low-to-high transition indicates a stop. Normal data bit transitions must occur within the low time of the clock period. These conditions are shown in Figure 4-1. The master generates the 7-bit slave address and the read/write (R/W) bit to open communication with another device and then wait for an acknowledge condition. The TAS5028 holds SDA low during acknowledge clock period to indicate an acknowledgement. When this occurs, the master transmits the next byte of the sequence. Each device is addressed by a unique 7-bit slave address plus R/W bit (1 byte). All compatible devices share the same signals via a bidirectional bus using a wired-AND connection. An external pullup resistor must be used for the SDA and SCL signals to set the high level for the bus.

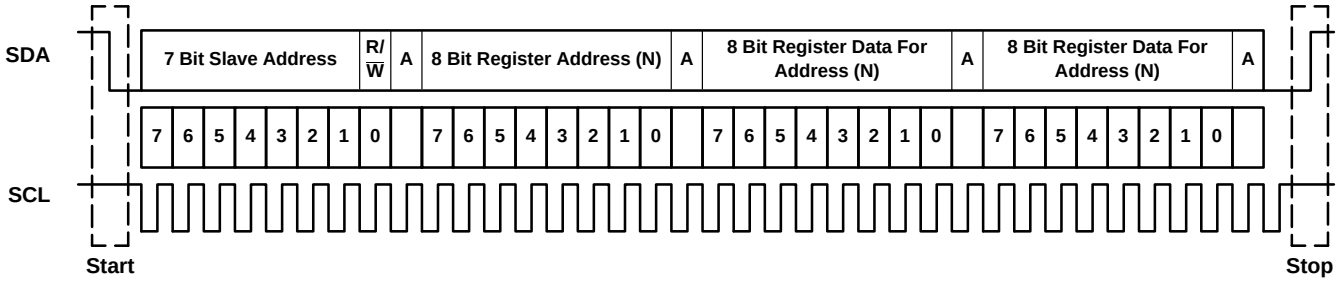


Figure 4-1. Typical I²C Sequence

There is no limit on the number of bytes that can be transmitted between start and stop conditions. When the last word transfers, the master generates a stop condition to release the bus. A generic data transfer sequence is shown in Figure 4-1.

The 7-bit address for the TAS5028 is 0011011.

4.2 Single and Multiple Byte Transfers

The serial control interface supports both single-byte and multiple-byte read / write operations for status registers and the general control registers associated with the PWM. However, for the DAP data processing registers, the serial control interface supports only multiple byte (4 bytes) read / write operations.

During multiple byte read operations, the TAS5028 responds with data, a byte at a time, starting at the subaddress assigned, as long as the master device continues to respond with acknowledges. If a particular subaddress does not contain 32 bits, the unused bits are read as logic 0.

During multiple byte write operations, the TAS5028 compares the number of bytes transmitted to the number of bytes that are required for each specific sub address. If a write command is received for a mixer coefficient, the TAS5028 expects to receive one 32-bit word. If fewer than 32 bits are required when a stop command (or another start command) is received, the data received is discarded.

Supplying a subaddress for each subaddress transaction is referred to as random I²C addressing. The TAS5028 also supports sequential I²C addressing. For write transactions, if a subaddress is issued followed by data for that subaddress and the fifteen subaddresses that follow, a sequential I²C write transaction has taken place, and the data for all 16 subaddresses is successfully received by the TAS5028. For I²C sequential write transactions, the subaddress then serves as the start address and the amount of data subsequently transmitted, before a stop or start is transmitted, determines how many subaddresses are written. As was true for random addressing, sequential addressing requires that a complete set of data be transmitted. If only a partial set of data is written to the last subaddress, the data for the last subaddress is discarded. However, all other data written is accepted; just the incomplete data is discarded.

4.3 Single Byte Write

As shown in Figure 4-2, a single byte data write transfer begins with the master device transmitting a start condition followed by the I²C device address and the read/write bit. The read/write bit determines the direction of the data transfer. For a write data transfer, the read/write bit will be a 0. After receiving the correct I²C device address and the read/write bit, the TAS5028 device responds with an acknowledge bit. Next, the master transmits the address byte or bytes corresponding to the TAS5028 internal memory address being accessed. After receiving the address byte, the TAS5028 again responds with an acknowledge bit. Next, the master device transmits the data byte to be written to the memory address being accessed. After receiving the data byte, the TAS5028 again responds with an acknowledge bit. Finally, the master device transmits a stop condition to complete the single byte data write transfer.

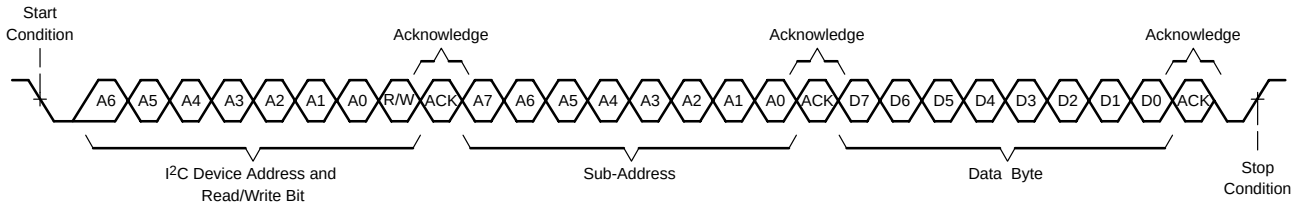


Figure 4-2. Single Byte Write Transfer

4.4 Multiple Byte Write

A multiple byte data write transfer is identical to a single byte data write transfer except that multiple data bytes are transmitted by the master device to TAS5028 as shown in Figure 4-3. After receiving each data byte, the TAS5028 responds with an acknowledge bit.

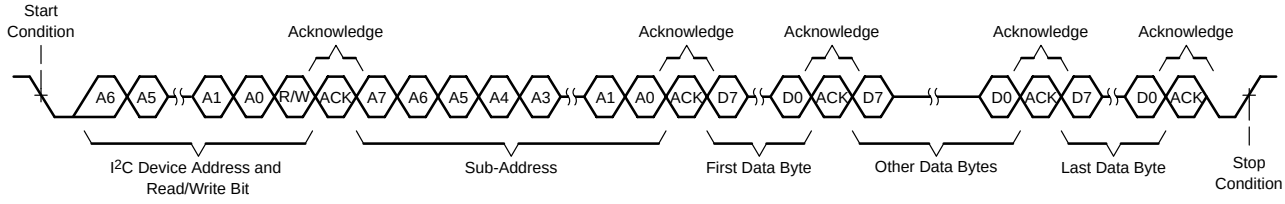


Figure 4-3. Multiple Byte Write Transfer

4.5 Incremental Multiple Byte Write

The I²C supports a special mode which permits I²C write operations to be broken up into multiple data write operations that are multiples of 4 data bytes. These are 6 byte, 10 byte, 14 byte, 18 byte, ... etc., write operations that are composed of a device address, read/write bit, and subaddress and any multiple of 4 bytes of data. This permits the system to incrementally write large register values without blocking other I²C transactions.

This feature is enabled by the append subaddress function in the TAS5028. This function enables the TAS5028 to append 4 bytes of data to a register that was opened by a previous I²C register write operation but has not received its complete number of data bytes. Since the length of the long registers is a multiple of 4 bytes, using 4-byte transfers will have only an integer number of append operations.

When the correct number of bytes has been received, the TAS5028 starts processing the data.

The procedure to perform an incremental multi-byte write operation is as follows:

1. Start a normal I²C write operation by sending the device address, write bit, register subaddress, and the first four bytes of the data to be written. At the end of that sequence, send a stop condition. At this point, the register has been opened and accepts the remaining data that is sent by writing 4-byte blocks of data to the append subaddress (0xFE).
2. At a later time, one or more append data transfers are performed to incrementally transfer the remaining number of bytes in sequential order to complete the register write operation. Each of these append operations will be composed of the device address, write bit, append subaddress (0xFE), and four bytes of data followed by a stop condition.
3. The operation will be terminated due to an error condition and the data will be flushed:
   a. If a new subaddress is written to the TAS5028 before the correct number of bytes have been written.
   b. If more or less than 4 bytes are data written at the beginning or during any of the append operations.
   c. If a read bit is sent.

4.6 Single Byte Read

As shown in Figure 4-4, a single byte data read transfer begins with the master device transmitting a start condition followed by the I²C device address and the read/write bit. For the data read transfer, both a write followed by a read are actually done. Initially, a write is done to transfer the address byte or bytes of the internal memory address to be read. As a result, the read/write bit will be a 0. After receiving the TAS5028 address and the read/write bit, the TAS5028 responds with an acknowledge bit. In addition, after sending the internal memory address byte or bytes, the master device transmits another start condition followed by the TAS5028 address and the read/write bit again. This time the read/write bit will be a 1, indicating a read transfer. After receiving the TAS5028 and the read/write bit the TAS5028 again responds with an acknowledge bit. Next, the TAS5028 transmits the data byte from the memory address being read. After receiving the data byte, the master device transmits a not acknowledge followed by a stop condition to complete the single byte data read transfer.

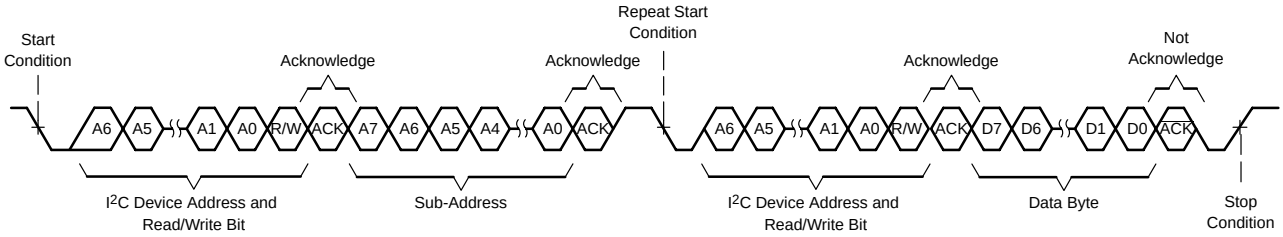


Figure 4-4. Single Byte Read Transfer

4.7 Multiple Byte Read

A multiple byte data read transfer is identical to a single byte data read transfer except that multiple data bytes are transmitted by the TAS5028 to the master device as shown in Figure 4-5. Except for the last data byte, the master device responds with an acknowledge bit after receiving each data byte.

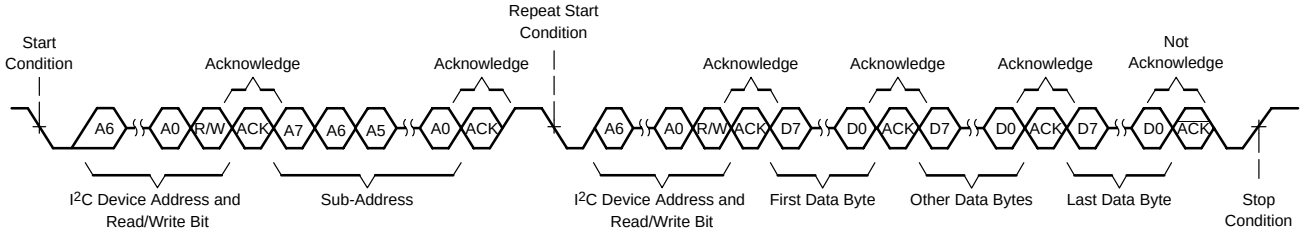


Figure 4-5. Multiple Byte Read Transfer

5 Serial Control I²C Register Summary

The TAS5028 slave address is 0x36. See the *Serial Control I_2C Register Bit Definitions* chapter for complete bit definitions.

Note that u indicates unused bits.

I²C SUBADDRESS	TOTAL BYTES	REGISTER FIELDS	DESCRIPTION OF CONTENTS	DEFAULT STATE
0x00	1	Clock control register	Set data rate and MCLK frequency	1. Fs = 48 kHz 2. MCLK = 256 Fs = 12.288 MHz
0x01	1	General status register	Clip indicator and ID code for the TAS5028	0x01
0x02	1	Error status register	PLL, SCLK, LRCLK, and frame slip errors	No errors
0x03	1	System control register 1	PWM high pass, clock set, un-mute select	1. PWM high pass disabled 2. Auto clock set 3. Hard un-mute on clock error recovery
0x04	1	System control register 2	Automute and de-emphasis control	1. Automute timeout disable 2. Post-DAP detection automute enabled 3. 8-Ch device input detection automute enabled 4. Un-mute threshold 6 dB over input 5. No de-emphasis
0x05 – 0x0C	1	Channel configuration registers	Configure channels 1, 2, 3, 4, 5, 6, 7, and 8	1. Enable backend reset 2. Valid low for reset 3. Valid low for mute 4. Normal BEPolarity 5. Don’t remap the output for the TAS5182 6. Don’t go low-low in mute 7. Don’t remap Hi-Z state to low-low state
0x0D	1	Headphone configuration register	Configure headphone output	1. Disable backend reset sequence 2 Valid does not have to be low for reset 3. Valid does not have to be low for mute 4. Normal BEPolarity 5. Don’t remap output to comply with 5182 6. Don’t go low-low in mute 7. Don’t remap Hi-Z state to low-low state
0x0E	1	Serial data interface register	Set serial data interface to right justified, I2S, or left justified	24-bit I2S
0x0F	1	Soft mute register	Soft mute for channels 1, 2, 3, 4, 5, 6, 7, and 8	Un-mute all channels
0x10 – 0x13			RESERVED	
0x14	1	Automute control	Set auto-mute delay and threshold	1. Set auto-mute delay = 5 ms 2. Set auto-mute threshold less than bit 8
0x15	1	Automute PWM threshold and backend reset period	Set PWM auto-mute threshold, set backend reset period	1. Set the PWM threshold the same as the TAS5028 input threshold 2. Set backend reset period = 5 ms
0x16	1	Modulation limit register	Set modulation index	97.7%
0x17-0x1A			RESERVED	

I²C SUBADDRESS	TOTAL BYTES	REGISTER FIELDS	DESCRIPTION OF CONTENTS	DEFAULT STATE
0x1B–0x22	1/Reg.	Inter-channel delay registers	Set inter-channel delay	Channel 1 delay = -23 DCLK periods Channel 2 delay = 0 DCLK periods Channel 3 delay = -16 DCLK periods Channel 4 delay = +16 DCLK periods Channel 5 delay = -24 DCLK periods Channel 6 delay = 8 DCLK periods Channel 7 delay = -8 DCLK periods Channel 8 delay = 24 DCLK periods
0x23	1	Inter-channel offset	Absolute delay offset for channel 1 (0 – 255)	Minimum absolute default = 0 DCLK periods
0x24-0x3F			RESERVED	
0x40	4	Bank switching command register	Set up DAP coefficients bank switching for banks 1, 2, and 3	Manual selection – Bank 1
0x41–0x48	32/Reg	See the *Input Mixer Registers (0x41 – 0x48, Channels 1 – 8)* section	8X8 input crossbar mixer setup	SDIN1-Left to input mixer 1 SDIN1-Right to input mixer 2 SDIN2-Left to input mixer 3 SDIN2-Right to input mixer 4 SDIN3-Left to input mixer 5 SDIN3-Right to input mixer 6 SDIN4-Left to input mixer 7 SDIN4-Right to input mixer 8
0x49–0x88			RESERVED	
0x89–0x90	8	Bass and Treble Bypass Ch 1 - 8	Bypass bass and treble for channels 1-8	Bass and treble bypassed for all channels
0x91–0xA9			RESERVED	
0xAA	8	sel op1-8 and mix to S	Select 0 to 2 of eight channels to output mixer S	Select channel 1 to PWM 1
0xAB	8	sel op1-8 and mix to T	Select 0 to 2 of eight channels to output mixer T	Select channel 2 to PWM 2
0xAC	8	sel op1-8 and mix to U	Select 0 to 2 of eight channels to output mixer U	Select channel 3 to PWM 3
0xAD	8	sel op1-8 and mix to V	Select 0 to 2 of eight channels to output mixer V	Select channel 4 to PWM 4
0xAE	8	sel op1-8 and mix to W	Select 0 to 2 of eight channels to output mixer W	Select channel 5 to PWM 5
0xAF	8	sel op1-8 and mix to X	Select 0 to 2 of eight channels to output mixer X	Select channel 6 to PWM 6
0xB0	12	sel op1-8 and mix to Y	Select 0 to 3 of eight channels to output mixer Y	Select channel 7 to PWM 7
0xB1	12	sel op1-8 and mix to Z	Select 0 to 3 of eight channels to output mixer Z	Select channel 8 to PWM 8
0xB2–0xCF			RESERVED	
0xD0	4	Vol, T and B slew rates	U (31:24), U (23:16), U (15:12) VSR(11:8), TBSR(7:0)	0x00, 0x00, 0x02, 0x3F
0xD1	4	Ch1 volume	Channel 1 volume	0 dB
0xD2	4	Ch2 volume	Channel 2 volume	0 dB
0xD3	4	Ch3 volume	Channel 3 volume	0 dB
0xD4	4	Ch4 volume	Channel 4 volume	0 dB

I²C SUBADDRESS	TOTAL BYTES	REGISTER FIELDS	DESCRIPTION OF CONTENTS	DEFAULT STATE
0xD5	4	Ch5 volume	Channel 5 volume	0 dB
0xD6	4	Ch6 volume	Channel 6 volume	0 dB
0xD7	4	Ch7 volume	Channel 7 volume	0 dB
0xD8	4	Ch8 volume	Channel 8 volume	0 dB
0xD9	4	Master volume	Master volume	Mute
OXDA	4	Bass filter set (1-5)	Bass filter set (all channels)	Filter set 3
0xDB	4	Bass filter index	Bass filter level (all channels)	0 dB
0xDC	4	Treble filter set (1-5)	Treble filter set (all channels)	Filter set 3
0xDD	4	Treble filter index	Treble filter level (all channels)	0 dB
0xDE	4	AM mode and tuned frequency register	Set-up AM mode for AM-interference reduction	AM mode disabled Select sequence 1 IF frequency = 455 kHz Use BCD-tuned frequency
0xDF			RESERVED	
0xE0	4	General control register	Six or eight channel configuration	Eight channel configuration
0xE1–0xFD			RESERVED	
0xFE	4 (min)	Multiple bit writer append register	Special register	N/A
0xFF			RESERVED	

6 Serial Control Interface Register Definitions

Unless otherwise noted, the I^2C register default values are in **bold** font.

Note that u indicates unused bits.

6.1 Clock Control Register (0x00)

Bit D1 is *Don't Care*.

Table 6-1. Clock Control Register

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	0	0	-	-	-		-	32-kHz data rate
0	0	1	-	-	-		-	38-kHz data rate
0	1	0	-	-	-		-	44.1-kHz data rate
0	**1**	**1**	**-**	**-**	**-**		**-**	**48-kHz data rate**
1	0	0	-	-	-		-	88.2-kHz data rate
1	0	1	-	-	-		-	96-kHz data rate
1	1	0	-	-	-		-	176.4-kHz data rate
1	1	1	-	-	-		-	192-kHz data rate
-	-	-	0	0	0			MCLK frequency = 64
-	-	-	0	0	1			MCLK frequency = 128
-	-	-	0	1	0			MCLK frequency = 192
-	**-**	**-**	**0**	**1**	**1**			**MCLK frequency = 256**
-	-	-	1	0	0			MCLK frequency = 384
-	-	-	1	0	1			MCLK frequency = 512
-	-	-	1	1	0			MCLK frequency = 768
-	-	-	1	1	1			Reserved
-	-	-	-	-	-		1	Clock register is valid (read only)
-	**-**	**-**	**-**	**-**	**-**		**0**	**Clock register is not valid (read only)**

6.2 General Status Register 0 (0x01)

Table 6-2. General Status Register (0x01)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
1	-	-	-	-	-	-	-	Clip indicator
-	1	-	-	-	-	-	-	Bank switching busy
-	**-**	**0**	**0**	**0**	**0**	**0**	**1**	**Identification code for TAS5028**

6.3 Error Status Register (0x02)

Note that the error bits are sticky bits that are not cleared by the hardware. This means that the software must clear the register (write zeroes) and than read them to determine if there are any persistent errors.

Table 6-3. Error Status Register (0X02)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
1	-	-	-	-	-	-	-	PLL phase lock error
-	1	-	-	-	-	-	-	PLL auto lock error
-	-	1	-	-	-	-	-	SCLK error
-	-	-	1	-	-	-	-	LRCLK error
-	-	-	-	1	-	-	-	Frame slip
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	No errors

6.4 System Control Register 1 (0x03)

Bit D5, D2, D1, and D0 are *Don't Care*.

Table 6-4. System Control Register 1

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	**-**		**-**	**-**				**PWM high pass disabled**
1	-		-	-				PWM high pass enabled
-	-		0					Soft unmute on recovery from clock error
-	**-**		**1**					**Hard unmute on recovery from clock error**

6.5 System Control Register 2 (0x04)

Bit D3 and D2 are *Don't Care*.

Table 6-5. System Control Register 2

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	**-**	**-**	**-**			**-**	**-**	**Reserved**
-	**0**	**-**	**-**			**-**	**-**	**PWM automute detection enabled**
-	1	-	-			-	-	PWM automute detection disabled
		0	**-**			**-**	**-**	**8 Ch device input detection automute enabled**
-	-	1	-			-	-	8 Ch device input detection automute disabled
-	**-**	**-**	**0**			**-**	**-**	**Unmute threshold 6 dB over input threshold**
-	-	-	1			-	-	Unmute threshold equal to input threshold
-	**-**	**-**	**-**			**0**	**0**	**No de-emphasis**
-	-	-	-			0	1	De-emphasis for Fs = 32 kHz
-	-	-	-			1	0	De-emphasis for Fs = 44.1 kHz
-	-	-	-			1	1	De-emphasis for Fs = 48 kHz

6.6 Channel Configuration Control Register (0x05-X0C)

Channels 1, 2, 3, 4, 5, 6, 7, and 8 are mapped into x05, x06, x07, x08, x09, x0A, x0B, and x0C.

Bit D0 is *Don't Care*.

Table 6-6. Channel Configuration Control Registers

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	-	-	-	-	-	-		Disable backend reset sequence for a channel - BEErrorRecEn
1	**-**	**-**	**-**	**-**	**-**	**-**		**Enable backend reset sequence for a channel**
-	0	-	-	-	-	-		Valid does not have to be low for this channel to be reset BEValidRst
-	**1**	**-**	**-**	**-**	**-**	**-**		**Valid must be low for this channel to be reset**
-	-	0	-	-	-	-		Valid does not have to be low for this channel to be muted BEValidMute
-	**-**	**1**	**-**	**-**	**-**	**-**		**Valid must be low for this channel to be muted**
-	**-**	**-**	**0**	**-**	**-**	**-**		**Normal BEPolarity**
-	-	-	1	-	-	-		Switches PWM+ and PWM- and invert audio signal
-	**-**	**-**	**-**	**0**	**-**	**-**		**Do not remap output to comply with 5182 interface**
-	-	-	-	1	-	-		Remap output to comply with 5182 interface
-	**-**	**-**	**-**	**-**	**0**	**-**		**Do not go to low low in mute - BELowMute**
-	-	-	-	-	1	-		Go to low-low in Mute
-	**-**	**-**	**-**	**-**	**-**	**0**		**Do not remap Hi-Z state to low-low state - BE5111BsMute**
-	-	-	-	-	-	1		Remap Hi-Z state to low-low state

6.7 Headphone Configuration Control Register (0x0D)

Bit D0 is *Don't Care*.

Table 6-7. Headphone Configuration Control Register

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	**-**	**-**	**-**	**-**	**-**	**-**		**Disable backend reset sequence for a channel - BEErrorRecEn**
1	-	-	-	-	-	-		Enable backend reset sequence for a channel
-	**0**	**-**	**-**	**-**	**-**	**-**		**Valid does not have to be low for this channel to be reset BEValidRst**
-	1	-	-	-	-	-		Valid must be low for this channel to be reset
-	**-**	**0**	**-**	**-**	**-**	**-**		**Valid does not have to be low for this channel to be muted BEValidMute**
-	-	1	-	-	-	-		Valid must be low for this channel to be muted
-	**-**	**-**	**0**	**-**	**-**	**-**		**Normal BEPolarity**
-	-	-	1	-	-	-		Switches PWM+ and PWM- and invert audio signal
-	**-**	**-**	**-**	**0**	**-**	**-**		**Do not remap output to comply with 5182 interface**
-	-	-	-	1	-	-		Remap output to comply with 5182 interface
-	**-**	**-**	**-**	**-**	**0**	**-**		**Do not go to low low in mute - BELowMute**
-	-	-	-	-	1	-		Go to low-low in Mute
-	**-**	**-**	**-**	**-**	**-**	**0**		**Do not remap Hi-Z state to low-low state - BE5111BsMute**
-	-	-	-	-	-	1		Remap Hi-Z state to low-low state

6.8 Serial Data Interface Control Register (0x0E)

Nine serial modes can be programmed I^2C.

Table 6-8. Serial Data Interface Control Register Format

RECEIVE SERIAL DATA INTERFACE FORMAT	WORD LENGTHS	D7-D4	D3	D2	D1	D0
Right justified	16	0000	0	0	0	0
Right justified	20	0000	0	0	0	1
Right justified	24	0000	0	0	1	0
I^2S	16	0000	0	0	1	1
I^2S	20	0000	0	1	0	0
I^2S	**24**	0000	**0**	**1**	**0**	**1**
Left justified	16	0000	0	1	1	0
Left justified	20	0000	0	1	1	1
Left justified	24	0000	1	0	0	0
Illegal		0000	1	0	0	1
Illegal		0000	1	0	1	0
Illegal		0000	1	0	1	1
Illegal		0000	1	1	0	0
Illegal		0000	1	1	0	1
Illegal		0000	1	1	1	0
Illegal		0000	1	1	1	1

6.9 Soft Mute Register (0x0F)

Table 6-9. Soft Mute Register

D7	D6	D5	D4	D3	D2	D1	D0	Function
-	-	-	-	-	-	-	1	Soft Mute Channel 1
-	-	-	-	-	-	1	-	Soft Mute Channel 2
-	-	-	-	-	1	-	-	Soft Mute Channel 3
-	-	-	-	1	-	-	-	Soft Mute Channel 4
-	-	-	1	-	-	-	-	Soft Mute Channel 5
-	-	1	-	-	-	-	-	Soft Mute Channel 6
-	1	-	-	-	-	-	-	Soft Mute Channel 7
1	-	-	-	-	-	-	-	Soft Mute Channel 8
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	**Unmute All Channels**

6.10 Automute Control Register(0x14)

Table 6-10. Automute Control Register

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
-	-	-	-	0	0	0	0	Set input automute and PWM automute delay to 1 ms
-	-	-	-	0	0	0	1	Set input automute and PWM automute delay to 2 ms
-	-	-	-	0	0	1	0	Set input automute and PWM automute delay to 3 ms
-	-	-	-	0	0	1	1	Set input automute and PWM automute delay to 4 ms
-	**-**	**-**	**-**	**0**	**1**	**0**	**0**	**Set input automute and PWM automute delay to 5 ms**
-	-	-	-	0	1	0	1	Set input automute and PWM automute delay to 10 ms
-	-	-	-	0	1	1	0	Set input automute and PWM automute delay to 20 ms
-	-	-	-	0	1	1	1	Set input automute and PWM automute delay to 30 ms
-	-	-	-	1	0	0	0	Set input automute and PWM automute delay to 40 ms
-	-	-	-	1	0	0	1	Set input automute and PWM automute delay to 50 ms
-	-	-	-	1	0	1	0	Set input automute and PWM automute delay to 60 ms
-	-	-	-	1	0	1	1	Set input automute and PWM automute delay to 70ms
-	-	-	-	1	1	0	0	Set input automute and PWM automute delay to 80 ms
-	-	-	-	1	1	0	1	Set input automute and PWM automute delay to 90 ms
-	-	-	-	1	1	1	0	Set input automute and PWM automute delay to 100 ms
-	-	-	-	1	1	1	1	Set input automute and PWM automute delay to 110 ms
0	0	0	0	-	-	-		Set input automute threshold less than Bit 1 (zero input signal), lowest automute threshold.
0	0	0	1	-	-	-	-	
0	0	1	0	-	-	-	-	Set input automute threshold less than Bit 2
0	0	1	1	-	-	-	-	Set input automute threshold less than Bit 3
0	1	0	0	-	-	-	-	Set input automute threshold less than Bit 4
0	1	0	1	-	-	-	-	Set input automute threshold less than Bit 5
0	1	1	0	-	-	-	-	Set input automute threshold less than Bit 6
0	1	1	1	-	-	-	-	Set input automute threshold less than Bit 7
1	**0**	**0**	**0**	**-**	**-**	**-**	**-**	**Set input automute threshold less than Bit 8**
1	0	0	1	-	-	-	-	Set input automute threshold less than Bit 9
1	0	1	0	-	-	-	-	Set input automute threshold less than Bit 10
1	0	1	1	-	-	-	-	Set input automute threshold less than Bit 11
1	1	0	0	-	-	-	-	Set input automute threshold less than Bit 12
1	1	0	1	-	-	-	-	Set input automute threshold less than Bit 13
1	1	1	0	-	-	-	-	Set input automute threshold less than Bit 14
1	1	1	1	-	-	-	-	Set input automute threshold less than Bit 15

6.11 Automute PWM Threshold and Backend Reset Period (0x15)

Table 6-11. Automute PWM Threshold and Backend Reset Period

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	**0**	**0**	**0**	**-**	**-**	**-**	**-**	**Set PWM automute threshold equals input automute threshold**
0	0	0	1	-	-	-	-	Set PWM automute threshold 1 bit more than input automute threshold
0	0	1	0	-	-	-	-	Set PWM automute threshold 2 bits more than input automute threshold
0	0	1	1	-	-	-	-	Set PWM automute threshold 3 bits more than input automute threshold
0	1	0	0	-	-	-	-	Set PWM automute threshold 4 bits more than input automute threshold
0	1	0	1	-	-	-	-	Set PWM automute threshold 5 bits more than input automute threshold
0	1	1	0	-	-	-	-	Set PWM automute threshold 6 bits more than input automute threshold
0	1	1	1	-	-	-	-	Set PWM automute threshold 7 bits more than input automute threshold
1	0	0	0	-	-	-	-	Set PWM automute threshold equals input automute threshold
1	0	0	1	-	-	-	-	Set PWM automute threshold 1 bit less than input automute threshold
1	0	1	0	-	-	-	-	Set PWM automute threshold 2 bits less than input automute threshold
1	0	1	1	-	-	-	-	Set PWM automute threshold 3 bits less than input automute threshold
1	1	0	0	-	-	-	-	Set PWM automute threshold 4 bits less than input automute threshold
1	1	0	1					Set PWM automute threshold 5 bits less than input automute threshold
1	1	1	0					Set PWM automute threshold 6 bits less than input automute threshold
1	1	1	1					Set PWM automute threshold 7 bits less than input automute threshold
-	-	-	-	0	0	0	0	Set backend reset period < 1 ms
-	-	-	-	0	0	0	1	Set backend reset period 1 ms
-	-	-	-	0	0	1	0	Set backend reset period 2 ms
-	-	-	-	0	0	1	1	Set backend reset period 3 ms
-	-	-	-	0	1	0	0	Set backend reset period 4 ms
-	**-**	**-**	**-**	**0**	**1**	**0**	**1**	**Set backend reset period 5 ms**
-	-	-	-	0	1	1	0	Set backend reset period 6 ms
-	-	-	-	0	1	1	1	Set backend reset period 7 ms
-	-	-	-	1	0	0	0	Set backend reset period 8 ms
-	-	-	-	1	0	0	1	Set backend reset period 9 ms
-	-	-	-	1	0	1	0	Set backend reset period 10 ms
-	-	-	-	1	0	1	1	Set backend reset period 10 ms
-	-	-	-	1	1	X	X	Set backend reset period 10 ms

6.12 Modulation Index Limit Register (0x16)

Table 6-12. Modulation Index Limit Register

D7	D6	D5	D4	D3	D2	D1	D0	LIMIT [DCLKS]	MIN WIDTH [DCLKS]	MODULATION INDEX
					0	0	0	1	2	99.2%
					0	0	1	2	4	98.4%
					0	**1**	**0**	**3**	**6**	**97.7%**
					0	1	1	4	8	96.9%
					1	0	0	5	10	96.1%
					1	0	1	6	12	95.3%
					1	1	0	7	14	94.5%
					1	1	1	8	16	93.8%

6.13 Interchannel Channel Delay Registers (0x1B - 0x22) and Offset Register (0x23)

Channels 1, 2, 3, 4, 5, 6, 7, and 8 are mapped into (0x1B, 0x1C, 0x1D, 0x1E, 0x1F, 0x20, 0x21, and 0x22).

Bits D1 and D0 are *Don't Care*.

Table 6-13. Interchannel Channel Delay Registers

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	**0**	**0**	**0**	**0**	**0**			**Minimum absolute delay, 0 DCLK cycles, default for channel 1**
0	1	1	1	1	1			Maximum positive delay, 31 x 4 DCLK cycles
1	0	0	0	0	0			Maximum negative delay, -32 x 4 DCLK cycles
1	**0**	**0**	**0**	**0**	**0**			**Default value for Channel 1 -32**
0	**0**	**0**	**0**	**0**	**0**			**Default value for Channel 2 0**
1	**1**	**0**	**0**	**0**	**0**			**Default value for Channel 3 -16**
0	**1**	**0**	**0**	**0**	**0**			**Default value for Channel 4 16**
1	**0**	**1**	**0**	**0**	**0**			**Default value for Channel 5 -24**
0	**0**	**1**	**0**	**0**	**0**			**Default value for Channel 6 8**
1	**1**	**1**	**0**	**0**	**0**			**Default value for Channel 7 -8**
0	**1**	**1**	**0**	**0**	**0**			**Default value for Channel 8 24**

The offset register is mapped into 0x23.

Table 6-14. Channel Offset Register

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	**Minimum absolute offset, 0 DCLK cycles default for channel 1**
1	1	1	1	1	1	1	1	Maximum absolute delay, 255 DCLK cycles

6.14 Bank Switching Command (0x40)

Bits D31-D24, D22-D19 are *Don't Care*.

Table 6-15. Bank Switching Command

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
								Unused bits

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
-					**0**	**0**	**0**	**Manual selection Bank 1**
-					0	0	1	Manual selection Bank 2
-					0	1	0	Manual selection Bank 3
-					0	1	1	Automatic bank selection
-					1	0	0	Update the values in Bank 1
-					1	0	1	Update the values in Bank 2
-					1	1	0	Update the values in Bank 3
0					1	1	1	Update only the bank map
0					x	x	x	Update the bank map using values in D15-D0
1					x	x	x	Do not update the bank map using values in D15-D0

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
1	**-**	**-**	**-**	**-**	**-**	**-**	**-**	**32-kHz data rate – Use Bank 1**
-	**1**	**-**	**-**	**-**	**-**	**-**	**-**	**38-kHz data rate – Use Bank 1**
-	**-**	**1**	**-**	**-**	**-**	**-**	**-**	**44.1-kHz data rate – Use Bank 1**
-	**-**	**-**	**1**	**-**	**-**	**-**	**-**	**48-kHz data rate – Use Bank 1**
-	**-**	**-**	**-**	**1**	**-**	**-**	**-**	**88.2-kHz data rate – Use Bank 1**
-	**-**	**-**	**-**	**-**	**1**	**-**	**-**	**96-kHz data rate – Use Bank 1**
-	**-**	**-**	**-**	**-**	**-**	**1**	**-**	**176.4-kHz data rate – Use Bank 1**
-	**-**	**-**	**-**	**-**	**-**	**-**	**1**	**192-kHz data rate – Use Bank 1**
1	**1**	**1**	**1**	**1**	**1**	**1**	**1**	**Default**

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
1	-	-	-	-	-	-	-	32-kHz data rate – Use Bank 2
-	1	-	-	-	-	-	-	38-kHz data rate – Use Bank 2
-	-	1	-	-	-	-	-	44.1-kHz data rate – Use Bank 2
-	-	-	1	-	-	-	-	48-kHz data rate – Use Bank 2
-	-	-	-	1	-	-	-	88.2-kHz data rate – Use Bank 2
-	-	-	-	-	1	-	-	96-kHz data rate – Use Bank 2
-	-	-	-	-	-	1	-	176.4-kHz data rate – Use Bank 2
-	-	-	-	-	-	-	1	192-kHz data rate – Use Bank 2
1	**1**	**1**	**1**	**1**	**1**	**1**	**1**	**Default**

6.15 Input Mixer Registers (0x41 – 0x48, Channels 1 - 8)

Input mixers 1, 2, 3, 4, 5, 6, 7, and 8 are mapped into registers 0x41, 0x42, 0x43, 0x44, 0x45, 0x46, 0x47, and 0x48.

Each gain coefficient is in 28-bit (5.23) format so 0x800000 is a gain of 1. Each gain coefficient is written as a 32-bit word with the upper 4 bits not used. For 8-gain coefficients, the total is 32 bytes.

Bold indicates the one channel that is passed through the mixer.

Table 6-16. Input Mixer Registers Format (0x41 – 0x48, Channels 1 - 8)

I^2C SUBADDRESS	TOTAL BYTES	REGISTER FIELDS	DESCRIPTION OF CONTENTS	DEFAULT STATE
0x41	32	**A_to_ipmix[1]**	**SDIN1-Left (Ch 1) A to Input Mixer 1 coefficient (default = 1) U (31:28), A_1 (27:24), A_1 (23:16), A_1 (15:8), A_1 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		B_to_ipmix[1]	SDIN1-Right (Ch 2) B to Input Mixer 1 coefficient (default = 0) U (31:28), B_1 (27:24), B_1 (23:16), B_1 (15:8), B_1 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[1]	SDIN2-Left (Ch 3) C to Input Mixer 1 coefficient (default = 0) U (31:28), C_1 (27:24), C_1 (23:16), C_1 (15:8), C_1 (7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[1]	SDIN2-Right (Ch 4) D to Input Mixer 1 coefficient (default = 0) U (31:28), D_1 (27:24), D_1 (23:16), D_1 (15:8), D_1 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[1]	SDIN3-Left (Ch 5) E to Input Mixer 1 coefficient (default = 0) U (31:28), E_1 (27:24), E_1 (23:16), E_1 (15:8), E_1 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[1]	SDIN3-Right (Ch 6) F to Input Mixer 1 coefficient (default = 0) U (31:28), F_1 (27:24), F_1 (23:16), F_1 (15:8), F_1 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[1]	SDIN4-Left (Ch 7) G to Input Mixer 1 coefficient (default = 0) U (31:28), G_1 (27:24), G_1 (23:16), G_1 (15:8), G_1 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[1]	SDIN4-Right (Ch 8) H to Input Mixer 1 coefficient (default = 0) U (31:28), H_1 (27:24), H_1 (23:16), H_1 (15:8), H_1 (7:0)	0x00, 0x00, 0x00, 0x00
0x42	32	A_to_ipmix[2]	SDIN1-Left (Ch 1) A to Input Mixer 2 coefficient (default = 0) U (31:28), A_2 (27:24), A_2 (23:16), A_2 (15:8), A_2 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[2]	**SDIN1-Right (Ch 2) B to Input Mixer 2 coefficient (default = 1) U (31:28), B_2 (27:24), B_2 (23:16), B_2 (15:8), B_2 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		C_to_ipmix[2]	SDIN2-Left (Ch 3) C to Input Mixer 2 coefficient (default = 0) U (31:28), C_2(27:24), C_2(23:16), C_2(15:8), C_2(7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[2]	SDIN2-Right (Ch 4) D to Input Mixer 2 coefficient (default = 0) U (31:28), D_2 (27:24), D_2 (23:16), D_2 (15:8), D_2 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[2]	SDIN3-Left (Ch 5) E to Input Mixer 2 coefficient (default = 0) U (31:28), E_2 (27:24), E_2 (23:16), E_2 (15:8), E_2 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[2]	SDIN3-Right (Ch 6) F to Input Mixer 2 coefficient (default = 0) U (31:28), F_2 (27:24), F_2 (23:16), F_2 (15:8), F_2 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[2]	SDIN4-Left (Ch 7) G to Input Mixer 2 coefficient (default = 0) U (31:28), G_2 (27:24), G_2 (23:16), G_2 (15:8), G_2 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[2]	SDIN4-Right (Ch 8) H to Input Mixer 2 coefficient (default = 0) U (31:28), H_2 (27:24), H_2 (23:16), H_2 (15:8), H_2 (7:0)	0x00, 0x00, 0x00, 0x00

I²C SUBADDRESS	TOTAL BYTES	REGISTER FIELDS	DESCRIPTION OF CONTENTS	DEFAULT STATE
0x43	32	A_to_ipmix[3]	SDIN1-Left (Ch 1) A to Input Mixer 3 coefficient (default = 0) U (31:28), A_3 (27:24), A_3 (23:16), A_3 (15:8), A_3 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[3]	SDIN1-Right (Ch 2) B to Input Mixer 3 coefficient (default = 0) U (31:28), B_3 (27:24), B_3 (23:16), B_3 (15:8), B_3 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[3]	**SDIN2-Left (Ch 3) C to Input Mixer 3 coefficient (default = 1) U (31:28), C_3 (27:24), C_3 (23:16), C_3 (15:8), C_3 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		D_to_ipmix[3]	SDIN2-Right (Ch 4) D to Input Mixer 3 coefficient (default = 0) U (31:28), D_3 (27:24), D_3 (23:16), D_3 (15:8), D_3 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[3]	SDIN3-Left (Ch 5) E to Input Mixer 3 coefficient (default = 0) U (31:28), E_3 (27:24), E_3 (23:16), E_3 (15:8), E_3 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[3]	SDIN3-Right (Ch 6) F to Input Mixer 3 coefficient (default = 0) U (31:28), F_3 (27:24), F_3 (23:16), F_3 (15:8), F_3 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[3]	SDIN4-Left (Ch 7) G to Input Mixer 3 coefficient (default = 0) U (31:28), G_3 (27:24), G_3 (23:16), G_3 (15:8), G_3 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[3]	SDIN4-Right (Ch 8) H to Input Mixer 3 coefficient (default = 0) U (31:28), H_3 (27:24), H_3 (23:16), H_3 (15:8), H_3 (7:0)	0x00, 0x00, 0x00, 0x00
0x44	32	A_to_ipmix[4]	SDIN1-Left (Ch 1) A to Input Mixer 4 coefficient (default = 0) U (31:28), A_4 (27:24), A_4 (23:16), A_4 (15:8), A_4 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[4]	SDIN1-Right (Ch 2) B to Input Mixer 4 coefficient (default = 0) U (31:28), B_4 (27:24), B_4 (23:16), B_4 (15:8), B_4 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[4]	SDIN2-Left (Ch 3) C to Input Mixer 4 coefficient (default = 0) U (31:28), C_4 (27:24), C_4 (23:16), C_4 (15:8), C_4 (7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[4]	**SDIN2-Right (Ch 4) D to Input Mixer 4 coefficient (default = 1) U (31:28), D_4 (27:24), D_4 (23:16), D_4 (15:8), D_4 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		E_to_ipmix[4]	SDIN3-Left (Ch 5) E to Input Mixer 4 coefficient (default = 0) U (31:28), E_4 (27:24), E_4 (23:16), E_4 (15:8), E_4 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[4]	SDIN3-Right (Ch 6) F to Input Mixer 4 coefficient (default = 0) U (31:28), F_4 (27:24), F_4 (23:16), F_4 (15:8), F_4 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[4]	SDIN4-Left (Ch 7) G to Input Mixer 4 coefficient (default = 0) U (31:28), G_4 (27:24), G_4 (23:16), G_4 (15:8), G_4 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[4]	SDIN4-Right (Ch 8) H to Input Mixer 4 coefficient (default = 0) U (31:28), H_4 (27:24), H_4 (23:16), H_4 (15:8), H_4 (7:0)	0x00, 0x00, 0x00, 0x00
0x45	32	A_to_ipmix[5]	SDIN1-Left (Ch 1) A to Input Mixer 5 coefficient (default = 0) U (31:28), A_5 (27:24), A_5 (23:16), A_5 (15:8), A_5 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[5]	SDIN1-Right (Ch 2) B to Input Mixer 5 coefficient (default = 0) U (31:28), B_5 (27:24), B_5 (23:16), B_5 (15:8), B_5 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[5]	SDIN2-Left (Ch 3) C to Input Mixer 5 coefficient (default = 0) U (31:28), C_5 (27:24), C_5 (23:16), C_5 (15:8), C_5 (7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[5]	SDIN2-Right (Ch 4) D to Input Mixer 5 coefficient (default = 0) U (31:28), D_5 (27:24), D_5 (23:16), D_5 (15:8), D_5 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[5]	**SDIN3-Left (Ch 5) E to Input Mixer 5 coefficient (default = 1) U (31:28), E_5 (27:24), E_5 (23:16), E_5 (15:8), E_5 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		F_to_ipmix[5]	SDIN3-Right (Ch 6) F to Input Mixer 5 coefficient (default = 0) U (31:28), F_5 (27:24), F_5 (23:16), F_5 (15:8), F_5 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[5]	SDIN4-Left (Ch 7) G to Input Mixer 5 coefficient (default = 0) U (31:28), G_5 (27:24), G_5 (23:16), G_5 (15:8), G_5 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[5]	SDIN4-Right (Ch 8) H to Input Mixer 5 coefficient (default = 0) U (31:28), H_5 (27:24), H_5 (23:16), H_5 (15:8), H_5 (7:0)	0x00, 0x00, 0x00, 0x00

I²C SUBADDRESS	TOTAL BYTES	REGISTER FIELDS	DESCRIPTION OF CONTENTS	DEFAULT STATE
0x46	32	A_to_ipmix[6]	SDIN1-Left (Ch 1) A to Input Mixer 6 coefficient (default = 0) U (31:28), A_6 (27:24), A_6 (23:16), A_6 (15:8), A_6 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[6]	SDIN1-Right (Ch 2) B to Input Mixer 6 coefficient (default = 0) U (31:28), B_6 (27:24), B_6 (23:16), B_6 (15:8), B_6 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[6]	SDIN2-Left (Ch 3) C to Input Mixer 6 coefficient (default = 0) U (31:28), C_6 (27:24), C_6 (23:16), C_6 (15:8), C_6 (7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[6]	SDIN2-Right (Ch 4) D to Input Mixer 6 coefficient (default = 0) U (31:28), D_6 (27:24), D_6 (23:16), D_6 (15:8), D_6 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[6]	SDIN3-Left (Ch 5) E to Input Mixer 6 coefficient (default = 0) U (31:28), E_6 (27:24), E_6 (23:16), E_6 (15:8), E_6 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[6]	**SDIN3-Right (Ch 6) F to Input Mixer 6 coefficient (default = 1) U (31:28), F_6 (27:24), F_6 (23:16), F_6 (15:8), F_6 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		G_to_ipmix[6]	SDIN4-Left (Ch 7) G to Input Mixer 6 coefficient (default = 0) U (31:28), G_6 (27:24), G_6 (23:16), G_6 (15:8), G_6 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[6]	SDIN4-Right (Ch 8) H to Input Mixer 6 coefficient (default = 0) U (31:28), H_6 (27:24), H_6 (23:16), H_6 (15:8), H_6 (7:0)	0x00, 0x00, 0x00, 0x00
0x47	32	A_to_ipmix[7]	SDIN1-Left (Ch 1) A to Input Mixer 7 coefficient (default = 0) U (31:28), A_7 (27:24), A_7 (23:16), A_7 (15:8), A_7 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[7]	SDIN1-Right (Ch 2) B to Input Mixer 7 coefficient (default = 0) U (31:28), B_7 (27:24), B_7 (23:16), B_7 (15:8), B_7 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[7]	SDIN2-Left (Ch 3) C to Input Mixer 7 coefficient (default = 0) U (31:28), C_7 (27:24), C_7 (23:16), C_7 (15:8), C_7 (7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[7]	SDIN2-Right (Ch 4) D to Input Mixer 7 coefficient (default = 0) U (31:28), D_7 (27:24), D_7 (23:16), D_7 (15:8), D_7 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[7]	SDIN3-Left (Ch 5) E to Input Mixer 7 coefficient (default = 0) U (31:28), E_7 (27:24), E_7 (23:16), E_7 (15:8), E_7 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[7]	SDIN3-Right (Ch 6) F to Input Mixer 7 coefficient (default = 0) U (31:28), F_7 (27:24), F_7 (23:16), F_7 (15:8), F_7 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[7]	**SDIN4-Left (Ch 7) G to Input Mixer 7 coefficient (default = 1) U (31:28), G_7 (27:24), G_7 (23:16), G_7 (15:8), G_7 (7:0)**	**0x00, 0x80, 0x00, 0x00**
		H_to_ipmix[7]	SDIN4-Right (Ch 8) H to Input Mixer 7 coefficient (default = 0) U (31:28), H_7 (27:24), H_7 (23:16), H_7 (15:8), H_7 (7:0)	0x00, 0x00, 0x00, 0x00
0x48	32	A_to_ipmix[8]	SDIN1-Left (Ch 1) A to Input Mixer 8 coefficient (default = 0) U (31:28), A_8 (27:24), A_8 (23:16), A_8 (15:8), A_8 (7:0)	0x00, 0x00, 0x00, 0x00
		B_to_ipmix[8]	SDIN1-Right (Ch 2) B to Input Mixer 8 coefficient (default = 0) U (31:28), B_8 (27:24), B_8 (23:16), B_8 (15:8), B_8 (7:0)	0x00, 0x00, 0x00, 0x00
		C_to_ipmix[8]	SDIN2-Left (Ch 3) C to Input Mixer 8 coefficient (default = 0) U (31:28), C_8 (27:24), C_8 (23:16), C_8 (15:8), C_8 (7:0)	0x00, 0x00, 0x00, 0x00
		D_to_ipmix[8]	SDIN2-Right (Ch 4) D to Input Mixer 8 coefficient (default = 0) U (31:28), D_8 (27:24), D_8 (23:16), D_8 (15:8), D_8 (7:0)	0x00, 0x00, 0x00, 0x00
		E_to_ipmix[8]	SDIN3-Left (Ch 5) E to Input Mixer 8 coefficient (default = 0) U (31:28), E_8 (27:24), E_8 (23:16), E_8 (15:8), E_8 (7:0)	0x00, 0x00, 0x00, 0x00
		F_to_ipmix[8]	SDIN3-Right (Ch 6) F to Input Mixer 8 coefficient (default = 0) U (31:28), F_8 (27:24), F_8 (23:16), F_8 (15:8), F_8 (7:0)	0x00, 0x00, 0x00, 0x00
		G_to_ipmix[8]	SDIN4-Left (Ch 7) G to Input Mixer 8 coefficient (default = 0) U (31:28), G_8 (27:24), G_8 (23:16), G_8 (15:8), G_8 (7:0)	0x00, 0x00, 0x00, 0x00
		H_to_ipmix[8]	**SDIN4-Right (Ch 8) H to Input Mixer 8 coefficient (default = 1) U (31:28), H_8 (27:24), H_8 (23:16), H_8 (15:8), H_8 (7:0)**	**0x00, 0x80, 0x00, 0x00**

6.16 Bass and Treble Bypass Register (0x89 – 0x90, Channels 1 - 8)

Channels 1, 2, 3, 4, 5, 6, 7, and 8 are mapped into registers 0x89, 0x8A, 0x8B, 0x8C,0x8D, 0x8E, 0x8F, and 0x90. Eight bytes are written for each channel. Each gain coefficient is in 28-bit (5.23) format so 0x800000 is a gain of 1. Each gain coefficient is written as a 32-bit word with the upper four bits not used.

Table 6-17. Bass and Treble Bypass Register Format (0x89-0x90)

REGISTER NAME	TOTAL BYTES	CONTENTS	INITIALIZATION VALUE
Channel bass and treble bypass	8	U 31:28), Bypass (27:24), Bypass (23:16), Bypass (15:8), Bypass (7:0)	0x00, 0x80, 0x00, 0x00
Channel bass and treble inline		U (31:28), Inline (27:24), Inline (23:16), Inline (15:8), Inline (7:0)	0x00, 0x00, 0x00, 0x00

6.17 8x2 Output Mixer Registers (0xAA – 0xAF)

Output mixers for channels 1–6 map to registers 0xAA – 0xAF.

Total data per register is 8 bytes.

Table 6-18. Output Mixer Control Register Format (Upper 4 Bytes)

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0					Select channel 1 to output mixer
0	0	0	1					Select channel 2 to output mixer
0	0	1	0					Select channel 3 to output mixer
0	0	1	1					Select channel 4 to output mixer
0	1	0	0					Select channel 5 to output mixer
0	1	0	1					Select channel 6 to output mixer
0	1	1	0					Select channel 7 to output mixer
0	1	1	1					Select channel 8 to output mixer
				G27	G26	G25	G24	Selected channel gain (upper 4 bits)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
G23	G22	G21	G20	G19	G18	G17	G16	Selected channel gain (continued)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
G15	G14	G13	G12	G11	G10	G9	G8	Selected channel gain (continued)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
G7	G6	G5	G4	G3	G2	G1	G0	Selected channel gain (lower 8 bits)

Table 6-19. Output Mixer Control (Lower 4 Bytes)

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0					Select channel 1 to output mixer
0	0	0	1					Select channel 2 to output mixer
0	0	1	0					Select channel 3 to output mixer
0	0	1	1					Select channel 4 to output mixer
0	1	0	0					Select channel 5 to output mixer
0	1	0	1					Select channel 6 to output mixer
0	1	1	0					Select channel 7 to output mixer
0	1	1	1					Select channel 8 to output mixer
				G27	G26	G25	G24	Selected channel gain (upper 4 bits)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
G23	G22	G21	G20	G19	G18	G17	G16	Selected channel gain (continued)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
G15	G14	G13	G12	G11	G10	G9	G8	Selected channel gain (continued)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
G7	G6	G5	G4	G3	G2	G1	G0	Selected channel gain (lower 8 bits)

6.18 8x3 Output Mixer Registers (0xB0 – 0xB1)

Output mixers for channels 7 and 8 map to registers 0xB0 and 0xB1.

Total data per register is 12 bytes.

Table 6-20. Output Mixer Control (Upper 4 Bytes)

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0					Select channel 1 to output mixer
0	0	0	1					Select channel 2 to output mixer
0	0	1	0					Select channel 3 to output mixer
0	0	1	1					Select channel 4 to output mixer
0	1	0	0					Select channel 5 to output mixer
0	1	0	1					Select channel 6 to output mixer
0	1	1	0					Select channel 7 to output mixer
0	1	1	1					Select channel 8 to output mixer
				G27	G26	G25	G24	Selected channel gain (upper 4 bits)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
G23	G22	G21	G20	G19	G18	G17	G16	Selected channel gain (continued)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
G15	G14	G13	G12	G11	G10	G9	G8	Selected channel gain (continued)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
G7	G6	G5	G4	G3	G2	G1	G0	Selected channel gain (lower 8 bits)

Table 6-21. Output Mixer Control (Middle 4 Bytes)

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0					Select channel 1 to output mixer
0	0	0	1					Select channel 2 to output mixer
0	0	1	0					Select channel 3 to output mixer
0	0	1	1					Select channel 4 to output mixer
0	1	0	0					Select channel 5 to output mixer
0	1	0	1					Select channel 6 to output mixer
0	1	1	0					Select channel 7 to output mixer
0	1	1	1					Select channel 8 to output mixer
				G27	G26	G25	G24	Selected channel gain (upper 4 bits)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
G23	G22	G21	G20	G19	G18	G17	G16	Selected channel gain (continued)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
G15	G14	G13	G12	G11	G10	G9	G8	Selected channel gain (continued)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
G7	G6	G5	G4	G3	G2	G1	G0	Selected channel gain (lower 8 bits)

Table 6-22. Output Mixer Control (Lower 4 Bytes)

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0					Select channel 1 to output mixer
0	0	0	1					Select channel 2 to output mixer
0	0	1	0					Select channel 3 to output mixer
0	0	1	1					Select channel 4 to output mixer
0	1	0	0					Select channel 5 to output mixer
0	1	0	1					Select channel 6 to output mixer
0	1	1	0					Select channel 7 to output mixer
0	1	1	1					Select channel 8 to output mixer
				G27	G26	G25	G24	Selected channel gain (upper 4 bits)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
G23	G22	G21	G20	G19	G18	G17	G16	Selected channel gain (continued)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
G15	G14	G13	G12	G11	G10	G9	G8	Selected channel gain (continued)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
G7	G6	G5	G4	G3	G2	G1	G0	Selected channel gain (lower 8 bits)

6.19 Volume Treble and Bass Slew Rates (0xD0)

Table 6-23. Volume Gain Update Rate (Slew Rate)

D31-D10	D9	D8	FUNCTION
0	0	0	512 step update at 4 Fs, 42.6 ms at 48 kHz
0	**0**	**1**	**1024 step update at 4 Fs, 85.3 ms at 48 kHz**
0	1	0	2048 step update at 4 Fs, 170 ms at 48 kHz
0	1	1	2048 step update at 4 Fs, 170 ms at 48 kHz

Table 6-24. Treble and Bass Gain Step Size (Slew Rate)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	0	0	0	0	0	0	0	No operation
0	0	0	0	0	0	1	1	
0	0	0	0	0	1	0	0	Minimum rate – Updates every 0.083 ms (every LRCLK at 48 kHz)
0	0	1	0	0	0	0	0	Update ever 0.67 ms (32 LRCLKs at 48 kHz)
0	**0**	**1**	**1**	**1**	**1**	**1**	**1**	**Default rate - Updates every 1.31 ms (63 LRCLKs at 48 kHz). This is the maximum constant time that can be set for all sample rates.**
1	1	1	1	1	1	1	1	Minimum rate – Updates every 5.08 ms (every 255 LRCLKs at 48 kHz)

6.20 Volume Registers (0xD1 - 0xD9)

Channels 1, 2, 3, 4, 5, 6, 7, and 8 are mapped into registers 0xD1, 0xD2, 0xD3, 0xD4, 0xD5, 0xD6, 0xD7, and 0xD8.

Master volume is mapped into register 0xD9.

Bits D31 - D12 are *Don't Care*.

Table 6-25. Volume Registers

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
								Unused bits

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
								Unused bits

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
				V11	**V10**	**V9**	**V8**	Volume

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
V7	**V6**	**V5**	**V4**	**V3**	**V2**	**V1**	**V0**	Volume

Table 6-26. Master and Individual Volume Controls

VOLUME INDEX (H)	GAIN/INDEX	EXPECTED	ACTUAL
001	17.75	17.81	17.81
002	17.5	17.56	17.56
003	17.25	17.31	17.31
004	17	17.06	17.06
005	16.75	16.81	16.81
006	16.5	16.56	16.56
007	16.25	16.31	16.31
008	16	16.05	16.05
009	15.75	15.8	15.8
00A	15.5	15.55	15.55
00B	15.25	15.3	15.3
00C	15	15.05	15.05
00D	14.75	14.8	14.8
00E	14.5	14.55	14.55
00F	14.25	14.3	14.3
010	14	14.05	14.05
044	1	1	1
045	0.75	0.75	0.75
046	0.5	0.5	0.5
047	0.25	0.25	0.25
048	0	0	0
049	-0.25	-0.25	-0.25
04A	-0.5	-0.5	-0.5
04B	-0.75	-0.75	-0.75
04C	-1	-1	-1
240	-126	-126.43	-126.43
241	-126.25	-126.68	-126.99
242	-126.5	-126.93	-126.99
243	-126.75	-127.19	-127.59
244	-127	-127.44	-127.59
245	**Mute**	**Mute**	**Mute**
TO			
3FF	**Mute**	**Mute**	**Mute**

6.21 Bass Filter Set Register (0xDA)

Bits D31-D27, D23-D19, D15-D11, and D7-D3 are *Don't Care*.

Table 6-27. Channel 8 Sub Woofer

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Bass filter set 1
0	0	0	0	0	0	1	0	Bass filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Bass filter set 3**
0	0	0	0	0	1	0	0	Bass filter set 4
0	0	0	0	0	1	0	1	Bass filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

Table 6-28. Channel 6 and 5 (Right and Left Lineout in Six Channel Configuration Right and Left Surround in Eight Channel Configuration)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Bass filter set 1
0	0	0	0	0	0	1	0	Bass filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Bass filter set 3**
0	0	0	0	0	1	0	0	Bass filter set 4
0	0	0	0	0	1	0	1	Bass filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

Table 6-29. Channel 4 and 3 (Right and Left Rear)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Bass filter set 1
0	0	0	0	0	0	1	0	Bass filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Bass filter set 3**
0	0	0	0	0	1	0	0	Bass filter set 4
0	0	0	0	0	1	0	1	Bass filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

Table 6-30. Channel 7, 2, 1 (Center, Right Front, and Left Front)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Bass filter set 1
0	0	0	0	0	0	1	0	Bass filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Bass filter set 3**
0	0	0	0	0	1	0	0	Bass filter set 4
0	0	0	0	0	1	0	1	Bass filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

6.22 Bass Filter Index Register (0xDB)

Index values above 0x24 are invalid.

Table 6-31. Bass Filter Index Register

I^2C SUBADDRESS	TOTAL BYTES	REGISTER NAME	DESCRIPTION OF CONTENTS	DEFAULT STATE
0xDB	Bass filter index (BFI)	4	Ch8_BFI (31:24), Ch65_BFI (23:16), Ch43_BFI (15:8), Ch721_BFI (7:0)	0x12, 0x12, 0x12, 0x12,

Table 6-32. Bass Filter Index Table

TREBLE INDEX VALUE	ADJUSTMENT (DB)	TREBLE INDEX VALUE	ADJUSTMENT (DB)
0x00	+18	0x13	-1
0x01	+17	0x14	-2
0x02	+16	0x15	-3
0x03	+15	0x16	-4
0x04	+14	0x17	-5
0x05	+13	0x18	-6
0x06	+12	0x19	-7
0x07	+11	0x1A	-8
0x08	+10	0x1B	-9
0x09	+9	0x1C	-10
0x0A	+8	0x1D	-11
0x0B	+7	0x1E	-12
0x0C	+6	0x1F	-13
0x0D	+5	0x20	-14
0x0E	+4	0x21	-15
0x0F	+3	0x22	-16
0x10	+2	0x23	-17
0x11	+1	0x24	-18
0x12	**0**		

6.23 Treble Filter Set Register (0xDC)

Bits D31-D27 are *Don't Care*.

Table 6-33. Channel 8 Sub Woofer

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Treble filter set 1
0	0	0	0	0	0	1	0	Treble filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Treble filter set 3**
0	0	0	0	0	1	0	0	Treble filter set 4
0	0	0	0	0	1	0	1	Treble filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

Table 6-34. Channel 6 and 5 (Right and Left Lineout in Six Channel Configuration or Right and Left Surround in Eight Channel Configuration)

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Treble filter set 1
0	0	0	0	0	0	1	0	Treble filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Treble filter set 3**
0	0	0	0	0	1	0	0	Treble filter set 4
0	0	0	0	0	1	0	1	Treble filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

Table 6-35. Channel 4 and 3 (Right and Left Rear)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Treble filter set 1
0	0	0	0	0	0	1	0	Treble filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Treble filter set 3**
0	0	0	0	0	1	0	0	Treble filter set 4
0	0	0	0	0	1	0	1	Treble filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

Table 6-36. Channel 7, 2, 1 (Center, Right Front, and Left Front)

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
0	0	0	0	0	0	0	0	No change
0	0	0	0	0	0	0	1	Treble filter set 1
0	0	0	0	0	0	1	0	Treble filter set 2
0	**0**	**0**	**0**	**0**	**1**	**1**	**1**	**Treble filter set 3**
0	0	0	0	0	1	0	0	Treble filter set 4
0	0	0	0	0	1	0	1	Treble filter set 5
0	0	0	0	0	1	1	0	Illegal
0	0	0	0	0	1	1	1	Illegal

6.24 Treble Filter Index (0xDD)

Index values above 0x24 are invalid.

Table 6-37. Treble Filter Index Register

I^2C SUBADDRESS	TOTAL BYTES	REGISTER NAME	DESCRIPTION OF CONTENTS	DEFAULT STATE
0xDD	Treble filter index (TFI)	4	Ch8_TFI (31:24), Ch65_TFI (23:16), Ch43_TFI (15:8), Ch721_TFI (7:0)	0x12,0x12,0x12,0x12

Table 6-38. Treble Filter Index

TREBLE INDEX VALUE	ADJUSTMENT (DB)	TREBLE INDEX VALUE	ADJUSTMENT (DB)
0x00	+18	0x13	-1
0x01	+17	0x14	-2
0x02	+16	0x15	-3
0x03	+15	0x16	-4
0x04	+14	0x17	-5
0x05	+13	0x18	-6
0x\06	+12	0x19	-7
0x07	+11	0x1A	-8
0x08	+10	0x1B	-9
0x09	+9	0x1C	-10
0x0A	+8	0x1D	-11
0x0B	+7	0x1E	-12
0x0C	+6	0x1F	-13
0x0D	+5	0x20	-14
0x0E	+4	0x21	-15
0x0F	+3	0x22	-16
0x10	+2	0x23	-17
0x11	+1	0x24	-18
0x12	**0**		

6.25 AM Mode Register (0xDE)

Bits D31-D21 are *Don't Care*.

Table 6-39. AM Mode Register

D31	D30	D29	D28	D27	D26	D25	D24	FUNCTION
								Unused bits

D23	D22	D21	D20	D19	D18	D17	D16	FUNCTION
			0	**-**	**-**	**-**	**-**	**AM mode disabled**
			1	-	-	-	-	AM mode enabled
			-	**0**	**0**	**-**	**-**	**Select sequence 1**
			-	0	1	-	-	Select sequence 2
			-	1	0	-	-	Select sequence 3
			-	1	1	-	-	Select sequence 4
			-	**-**	**-**	**0**	**-**	**IF frequency 455**
			-	-	-	1	-	IF frequency 262.5
			-	**-**	**-**	**-**	**0**	**Use BCD tuned frequency**
			-	-	-	-	1	Use binary tuned frequency

Table 6-40. AM Tuned Frequency Register in BCD Mode (Lower 2 Bytes of 0xDE)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
0	0	0	B0	-	-	-	-	BCD frequency (1000s kHz)
-	-	-	-	B3	B2	B1	B0	BCD frequency (100s kHz)
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	**Default value**

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
B3	B2	B1	B0	-	-	-	-	BCD frequency (10s kHz)
-	-	-	-	B3	B2	B1	B0	BCD frequency (1s kHz)
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	**Default value**

Table 6-41. AM Tuned Frequency Register in Binary Mode (Lower 2 Bytes of 0xDE)

D15	D14	D13	D12	D11	D10	D9	D8	FUNCTION
0	0	0	0	0	B10	B9	B8	Binary frequency (upper 3 bits)
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	**Default value**

D7	D6	D5	D4	D3	D2	D1	D0	FUNCTION
B7	B6	B5	B4	B3	B2	B1	B0	Binary frequency (lower 8 bits)
0	**0**	**0**	**0**	**0**	**0**	**0**	**0**	**Default value**

6.26 General Control Register (0xE0)

Bits D31-D4 are zero. Bit D0 is *Don't Care*.

Table 6-42. General Control Register

D31 – D4	D3	D2	D1	D0	FUNCTION
0		**-**	**0**		**/8 channel configuration**
0		-	1		6 channel configuration

6.27 Incremental Multiple Write Append Register (0xFE)

This is a special register used to append data to a previously opened register.

7 TAS5028 Example Application Schematic

The following page contains an example application schematic for the TAS5028.

TAS5028 Example Application Schematic
(Circuit is Subject To Change Without Notice)

Left + Right Line Out

/LINE_OUT_ENABLE
LINE OUTPUT
Phono socket J950
Phono socket J951
+3.3V +5.0V
+5.0V
+3.3V
/OE
OUT_R
OUT_L
PWM_P_R
PWM_M_R
PWM_M_L
PWM_P_L
GND
2 Channel Line Out (TLV272)

Left + Right Headphone

HEADPHONE OUTPUT
J900
Mini-Jack (3.5mm)
+5.0V
+5.0V
OUT_R
OUT_GND
OUT_L
PWM_HPM_L
PWM_HPP_L
PWM_HPM_R
PWM_HPP_R
2 Channel Headphone Design (TPA112)

R20 22.0R +3.3V
C25 220nF
C26 10uF
GND

MCLK

C10 10nF
R10 200R
R11 200R
C13 10nF
C11 100nF
C12 100nF
C14 100nF
C29 100nF

+3.3V R12 2R
C15 100nF
C16 10uF
C17 100nF
C18 1nF
R13 3.30R
GND

/RESET_TAS5028
/HP_SEL
/PDN_TAS5028
/MUTE_TAS5028

+3.3V
R14 1R
C19 10uF
C20 100nF
C21 100nF
GND

X10
13.5MHz
R21 1M
C27 15pF
C28 15pF

U10
TAS5028

RESERVED MCLK PWM_HPPR PWM_HPMR PWM_HPPL PWM_HPML PWM_P_6 PWM_M_6 PWM_P_5 PWM_M_5 DVDD_PWM DVSS_PWM PWM_P_8 PWM_M_8 PWM_P_7 PWM_M_7
VRA_PLL PLL_FLT_RET PLL_FLTM PLL_FLTP AVSS AVSS VRD_PLL AVSS_PLL AVDD_PLL VBGAP RESET HP_SEL PDN MUTE DVDD DVSS
VR_DPLL OSC_CAP XTL_OUT XTL_IN RESERVED RESERVED RESERVED SDA SCL LRCLK SCLK SDIN4 SDIN3 SDIN2 SDIN1 RESERVED
VR_PWM PWM_P_4 PWM_M_4 PWM_P_3 PWM_M_3 PWM_P_2 PWM_M_2 PWM_P_1 PWM_M_1 VAILD DVSS BKND_ERR DVDD DVSS DVSS VR_DIG

/VALID
/BKND_ERR_TAS5028
+3.3V
R18 1R
C22 100nF
C23 10uF
C24 100nF
GND

SDIN1
SDIN2
SDIN3
SDIN4
SCLK
LRCLK
SCL
SDA

PSU and Interface Logic

V-HBRIDGE GVDD +5.0V +3.3V
PSVC_MCPU
PSVC_TAS5028
CONF_SEL
/VALID
/VALID_CH5+CH6
/LINE_OUT_ENABLE
/RESET
/RESET_TAS5028
/BKND_ERR
/BKND_ERR_TAS5028
/SD1_TAS5121
/SD1
/SD2_TAS5121
/SD2
/OTW_TAS5121
/OTW

CH6 TAS5121 H-Bridge Output Stage

/SD2_TAS5121
/OTW_TAS5121
/VALID_CH5+CH6
/SD1_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
RIGHT BACK SURROUND SPEAKER OUTPUT
J600

CH5 TAS5121 H-Bridge Output Stage

/SD2_TAS5121
/OTW_TAS5121
/VALID_CH5+CH6
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
LEFT BACK SURROUND SPEAKER OUTPUT
J500

CH8 TAS5121 H-Bridge Output Stage

/SD2_TAS5121
/OTW_TAS5121
/VALID
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
SUBWOOFER SPEAKER OUTPUT
J800

CH7 TAS5121 H-Bridge Output Stage

/SD1_TAS5121
/OTW_TAS5121
/VALID
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
CENTER SPEAKER OUTPUT
J700

CH4 TAS5121 H-Bridge Output Stage

/SD2_TAS5121
/OTW_TAS5121
/VALID
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
RIGHT SURROUND SPEAKER OUTPUT
J400

CH3 TAS5121 H-Bridge Output Stage

/SD2_TAS5121
/OTW_TAS5121
/VALID
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
LEFT SURROUND SPEAKER OUTPUT
J300

CH2 TAS5121 H-Bridge Output Stage

/SD1_TAS5121
/OTW_TAS5121
/VALID
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
RIGHT SPEAKER OUTPUT
J200

CH1 TAS5121 H-Bridge Output Stage

/SD1_TAS5121
/OTW_TAS5121
/VALID
/SHUTDOWN_TAS5121
/TEMP_WARNING
PWM_P
PWM_M
/VALID
V-HBRIDGE
GVDD
OUT_1
OUT_2
LEFT SPEAKER OUTPUT
J100

PACKAGING INFORMATION

Orderable Device	Status (1)	Package Type	Package Drawing	Pins	Package Qty	Eco Plan (2)	Lead/Ball Finish	MSL Peak Temp (3)
TAS5028PAG	ACTIVE	TQFP	PAG	64	160	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-4-260C-72 HR
TAS5028PAGG4	ACTIVE	TQFP	PAG	64	160	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-4-260C-72 HR
TAS5028PAGR	ACTIVE	TQFP	PAG	64	1500	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-4-260C-72 HR
TAS5028PAGRG4	ACTIVE	TQFP	PAG	64	1500	Green (RoHS & no Sb/Br)	CU NIPDAU	Level-4-260C-72 HR

(1) The marketing status values are defined as follows:
ACTIVE: Product device recommended for new designs.
LIFEBUY: TI has announced that the device will be discontinued, and a lifetime-buy period is in effect.
NRND: Not recommended for new designs. Device is in production to support existing customers, but TI does not recommend using this part in a new design.
PREVIEW: Device has been announced but is not in production. Samples may or may not be available.
OBSOLETE: TI has discontinued the production of the device.

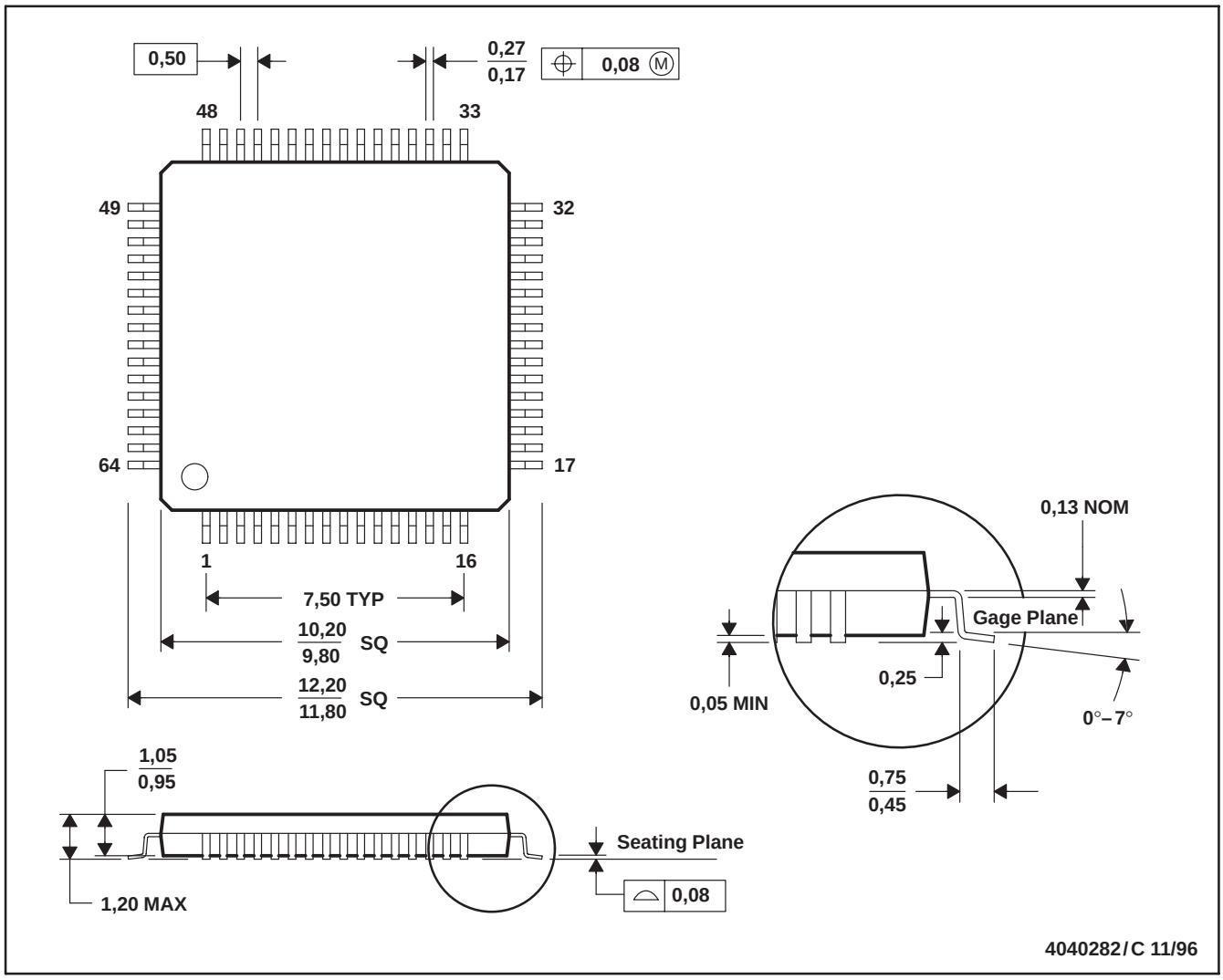
(2) Eco Plan - The planned eco-friendly classification: Pb-Free (RoHS), Pb-Free (RoHS Exempt), or Green (RoHS & no Sb/Br) - please check http://www.ti.com/productcontent for the latest availability information and additional product content details.
TBD: The Pb-Free/Green conversion plan has not been defined.
Pb-Free (RoHS): TI's terms "Lead-Free" or "Pb-Free" mean semiconductor products that are compatible with the current RoHS requirements for all 6 substances, including the requirement that lead not exceed 0.1% by weight in homogeneous materials. Where designed to be soldered at high temperatures, TI Pb-Free products are suitable for use in specified lead-free processes.
Pb-Free (RoHS Exempt): This component has a RoHS exemption for either 1) lead-based flip-chip solder bumps used between the die and package, or 2) lead-based die adhesive used between the die and leadframe. The component is otherwise considered Pb-Free (RoHS compatible) as defined above.
Green (RoHS & no Sb/Br): TI defines "Green" to mean Pb-Free (RoHS compatible), and free of Bromine (Br) and Antimony (Sb) based flame retardants (Br or Sb do not exceed 0.1% by weight in homogeneous material)

(3) MSL, Peak Temp. -- The Moisture Sensitivity Level rating according to the JEDEC industry standard classifications, and peak solder temperature.

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PAG (S-PQFP-G64) **PLASTIC QUAD FLATPACK**



NOTES: A. All linear dimensions are in millimeters.
B. This drawing is subject to change without notice.
C. Falls within JEDEC MS-026