Integrating Mixed-Signal Solutions

PRELIMINARY DATA SHEET

STAC9460/62

Two and Six-Channel, 24-Bit, 192 kHz Audio Codec

PRELIMINARY INFORMATION 8/24/01

2-9460-D1-1-0-0801

STAC9460/62

Two and Six-Channel, 24-Bit, 192 kHz Audio Codec

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2. PRODUCT BRIEF

SigmaTel's STAC9460/62 are six and two-channel general-purpose 24-bit, full duplex, audio codecs for use in consumer applications. The STAC9460/62 incorporate SigmaTel's proprietary Sigma-Delta technology to achieve ADC and DAC SNRs in excess of 100 dB. The DACs, and ADCs are integrated with analog I/Os, which include two differential analog and two MIC inputs. There are three audio I^2S inputs and an I^2S digital output. The STAC9460/62 communicates via a standard two-wire serial interface providing simplicity in the audio system design. Packaged in a 28-pin SSOP, the STAC9460 and STAC9462 require minimal PCB space for implementation.

The STAC9460 provides variable sample rate D-A and A-D conversion, as well as analog processing. Supported DAC audio sample rates include 32 kHz, 44.1 kHz, 48 kHz, 88.2 kHz, 96 kHz, 176.4 kHz and 192 kHz. Supported ADC audio sample rates inclued 32 kHz, 44.1 kHz, 48 kHz, 88.2 kHz,and 96 kHz. The digital data interface communicates via a standard $I^2C^®$ compatible serial control interface and I^2S digital audio interface. The stereo sample rate ADC's provide record capability from the microphone inputs or the differential inputs. All DAC's operate at 24-bit resolution with the sample rate based on the MCLK and programmable registers. The ADCs operate at 24-bit resolution and supply full 24-bit filter data.

The STAC9460 supports three I^2S digital audio inputs and an I^2S digital audio output. These digital I/O options provide for a number of advanced architectural implementations, with volume controls and mute capabilities built directly into the codec for each individual channel. The output volume ranges from 0 dB to -95 dB with .75 dB steps. For MIC input, the input volume ranges from 0 dB to 22.5 dB with 1.5 dB steps. The STAC9460 also supports a single-line format.

The STAC9460 is designed primarily to support 6-channel audio. True AC-3 playback can be achieved for 6-speaker applications by taking advantage of the STAC9460 architecture and combining it with the appropriate processing. This product is ideal for home theatre, DVD, karaoke, and set-top-box applications.

2.1. FEATURES

- High performance ΣΔ technology
- Two or six channels with independent volume controls
- 24-bit full duplex stereo DACs
- 24-bit full duplex stereo ADC
- 32, 44.1, 48, 88.2, 96, 176.4 and 192 kHz DAC sample rates
- 32, 44.1, 48, 88.2 and 96 kHz ADC sample rates
- Standard $I^2C^®$ compatible and I^2S serial interfaces
- Digital de-emphasis capability
- DAC and ADC SNR >100 dB
- Differential Stereo Analog Input
- Dual mic inputs with independent volume controls
- 28-pin SSOP package
- Energy saving dynamic power modes
- 5V Analog with 3.3V or 5V Digital capability

2.2. Ordering Information

PART NUMBER	CHANNELS	PACKAGE	TEMPERATURE RANGE	SUPPLY RANGE
STAC9460S	6 DAC; 2 ADC	28-pin SSOP	0 °C to +70 °C	AVdd = 5V, DVdd = 3.3V or 5V
STAC9462S	2 DAC; 2 ADC	28-pin SSOP	0 °C to +70 °C	AVdd = 5V, DVdd = 3.3V or 5V
STEECBAC9460B (Evaluation Board): please send email request to apps@sigmatel.com				

Note: SigmaTel reserves the right to change specifications without notice.

2.3. Block Diagram

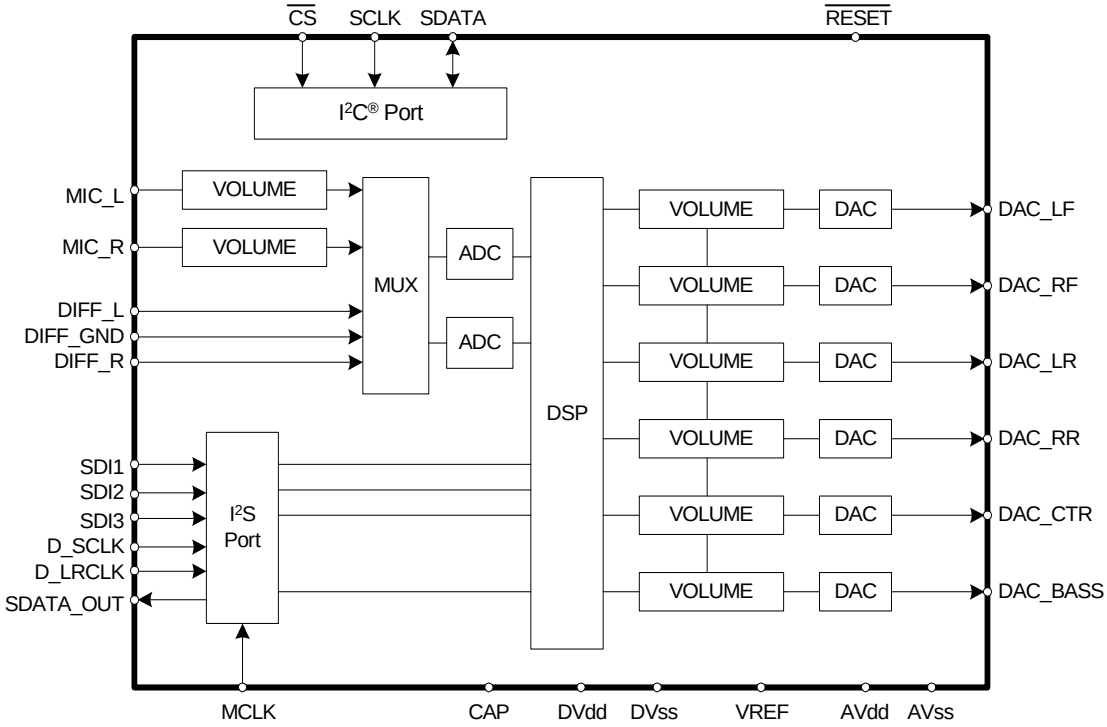

Figure 1. STAC9460 Block Diagram

2.4. Related Materials

- Product Brief
- Evaluation Boards
- Reference Designs

2.5. Additional Support

Additional product and company information can be obtained by going to the SigmaTel website at: ***www.sigmatel.com***

3. CHARACTERISTICS AND SPECIFICATIONS

3.1. Absolute Maximum Ratings

Voltage on any pin relative to Ground	Vss - 0.3 V TO Vdd + 0.3 V
Operating Temperature	0 ºC TO 70 ºC
Storage Temperature	-55 ºC TO +125 ºC
Soldering Temperature	260 ºC FOR 10 SECONDS
Output Current per Pin	± 4 mA

3.2. Recommended Operating Conditions

PARAMETER		MIN	TYP	MAX	UNITS
Power Supplies	+ 3.3 V Digital	3.135	3.3	3.435	V
	+ 5 V Digital	4.75	5	5.25	V
	+ 5 V Analog	4.75	5	5.25	V
Ambient Temperature		0	-	70	ºC

3.3. Power Consumption

PARAMETER		MIN	TYP	MAX	UNITS
Digital Supply Current	+ 5 V Digital	-	50	-	mA
	+ 3.3 V Digital	-	30	-	mA
Analog Supply Current	+ 5 V Analog	-	70	-	mA

3.4. Static Digital Specifications

($T_{ambient}$ = 25 ºC, DVdd = 5.0 V or 3.3 V ± 5%, AVss=DVss=0 V; 50pF external load)

PARAMETER	SYMBOL	MIN	TYP	MAX	UNITS
Input Voltage Range	Vin	-0.30		DVdd + 0.30	V
Low level input range	Vil	-	-	0.30xDVdd	V
High level input voltage	Vih	0.65xDVdd	-	-	V
High level output voltage	Voh	0.90xDVdd	-	-	V
Low level output voltage	Vol	-	-	0.2xDVdd	V
Input Leakage Current (Digital inputs)	-	-10	-	10	uA
Output Leakage Current (Digital outputs)	-	-10	-	10	uA
Output buffer drive current	-	-	4	-	mA

3.5. STAC9460 Analog Performance Characteristics

($T_{ambient}$ = 25 °C, AVdd = 5.0V ± 5%, DVdd = 5.0V ± 5%, AVss=DVss=0V; 1 kHz input sine wave; Sample Frequency = 48 kHz; 0 dB = 1 Vrms, 10 KΩ/50 pF load, Testbench Characterization BW: 20 Hz – 20 kHz, 0 dB settings on all gain stages)

PARAMETER	MIN	TYP	MAX	UNITS
Full Scale Input Voltage:				
Differential Inputs	-	1.0	-	Vrms
Mic Inputs	-	1.0	-	Vrms
Full Scale Output Voltage: Line Output 5V	-	1.0	-	Vrms
Analog Frequency Response (Note 1)	20	-	20,000	Hz
Digital S/N (Note 2)				
D/A 5V	-	104	-	dB
A/D 5V	-	100	-	dB
Total Harmonic Distortion: Line Output (Note 3)	-	90	-	dB
D/A Frequency Response (Note 4)	20	-	20,000	Hz
A/D Frequency Response (Note 4)	20	-	20,000	Hz
Transition Band(Note 6)	19,200	-	28,800	Hz
Stop Band(Note 6)	28,800	-	-	Hz
Deviation from Linear Phase (Note 6)	-	-	1	degree
Stop Band Rejection	-85	-	-	dB
Out-of-Band Rejection (Note 7)	-	-40	-	dB
Group Delay	-	-	1	ms
Power Supply Rejection Ratio (1kHz)	-	-40	-	dB
Crosstalk between Input channels	-	-	-85	dB
Spurious Tone Rejection	-	-100	-	dB
Mic Gain Step Size	-	1.5	-	dB
Mic Input Impedance	-	40	-	KΩ
Differential Input Impedance	-	18		KΩ
Input Capacitance	-	15	-	pF
VREFout	-	0.45 x AVdd	-	V
Interchannel Gain Mismatch ADC	-	-	0.5	dB
Interchannel Gain Mismatch DAC	-	-	0.5	dB
Gain Drift	-	100	-	ppm/ºC
DAC Offset Voltage	-	10	50	mV
External Load Impedance	10	-	-	KΩ
Mute Attenuation (Vrms input)	90	96	-	dB

Note:
1. ± 1 dB limits. If the sample rate is greater than or equal to 96 kHz the max frequency response becomes 40 kHz.
2. The ratio of the rms output level with 1 kHz full scale input to the rms output level with all zeros into the digital input. Measured “A weighted” over a 20 Hz to a 20 kHz bandwidth. (AES17-1991 Idle Channel Noise or EIAJ CP-307 Signal-to-noise Ratio).
3. 0 dB gain, 20 kHz BW, 48 kHz Sample Frequency
4. ± 0.25 dB limits. The D/A freq. response becomes 40 kHz with sampling rates > 96 kHz. At ±3dB the response range is from 20-22,500Hz at 48kHz, or 20-20,000Hz @ 44.1kHz or 20-45,000Hz @ 96kHz.
5. Transition band is 40-60% of sample rate. Stop band begins at 60% of sample rate.
6. Digital De-Emphasis OFF.
7. The integrated Out-of-Band noise generated by the DAC process, during normal PCM audio playback, over a bandwidth 28.8 to 100 kHz, with respect to a 1 Vrms DAC output.

4. TYPICAL CONNECTION DIAGRAM

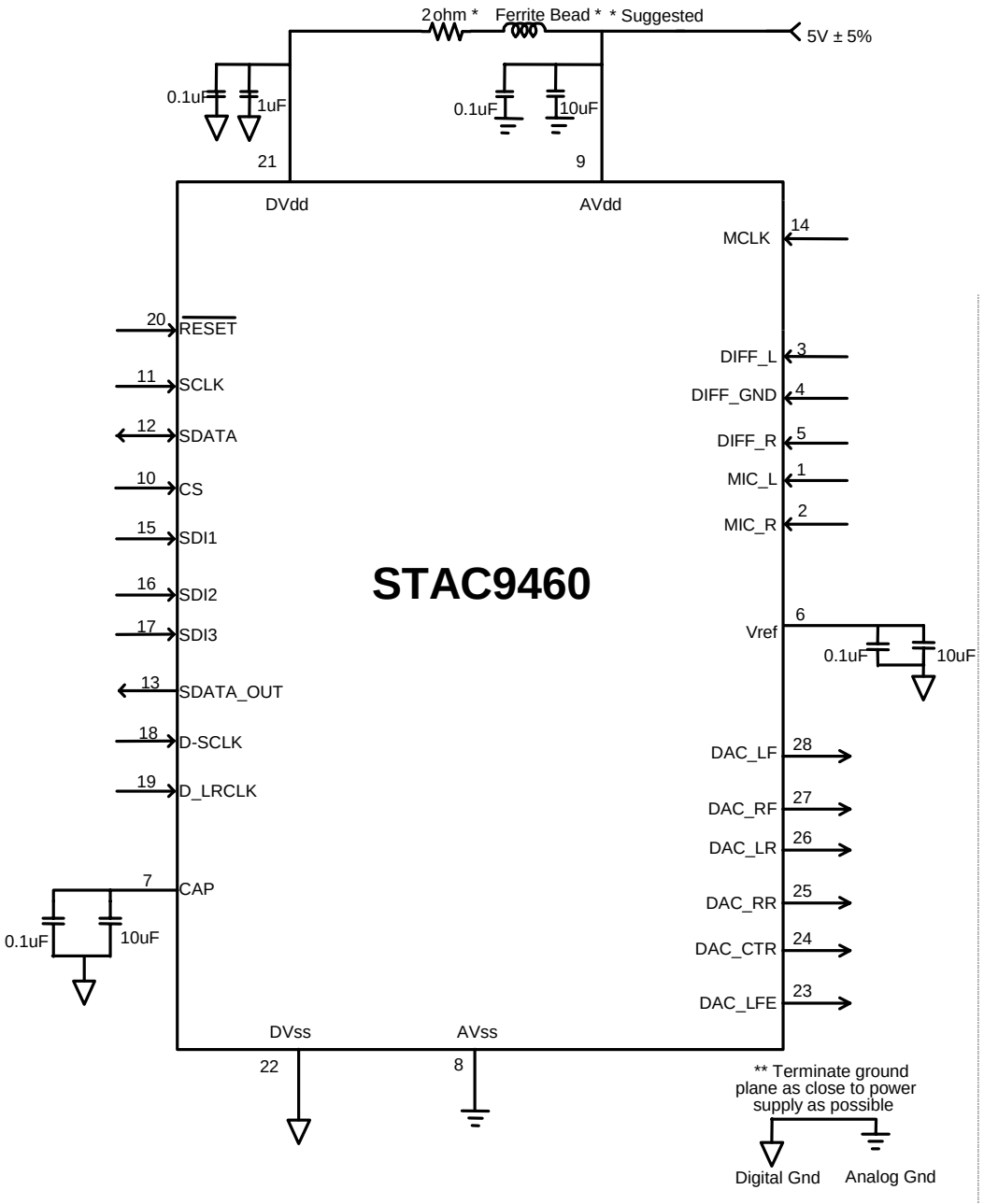

Figure 2. Typical Connection Diagram

5. SERIAL INTERFACE

Below is the figure for the serial interface between the **STAC9460/62** and a µC or µP. All register settings and chip control are performed via this serial interface, except for the address LSB.

Note: This functions as a standard 2-wire I^2C compatible interface, however, the CS (chip select) line offers address flexibility and must be either hard wired to ground or tied to Vdd if no other chips are connected to the bus. Refer to Table 7.1.1 on page 14 for additional information.

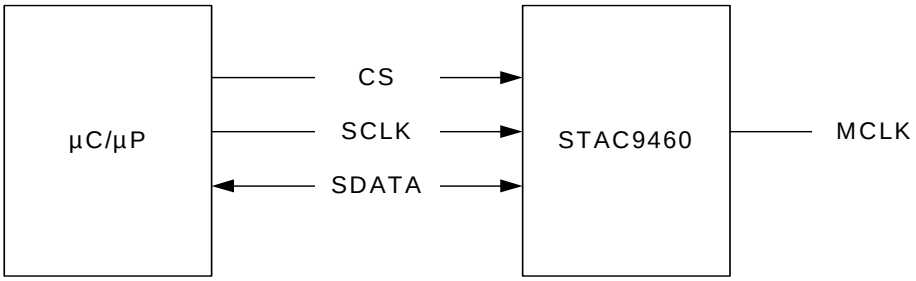

Figure 3. Serial interface to microcontroller or microprocessor

5.1. Clocking

The STAC9460/62 derives its clock from an externally connected clock through the MCLK pin in combination with the Master CLocking Register, which is further explained in section 7.1.9.

5.2. Reset

There are two types of resets as detailed below:

- A hard reset is achieved by driving the reset line low
- A soft reset is achieved by writing to the Reset/Status register (00h)

By writing to the Reset/Status Register (00h) a reset for the Address Control Register will occur. Writing any value to this register performs a register reset, which causes all registers to revert to their default values. This soft reset will also place the I^2C state machine in a "stop" condition, and will not continue to auto-increment through the address space. Additional information about the Address Control Register can be found in section 7.1.12.

6. DIGITAL AUDIO INTERFACE

6.1. I²S Serial Interface

The **STAC9460/62** communicates digital audio information through an I²S digital serial interface. The I²S interface can be configured with the standard I²S format, left justified, right justified or one line format. Input signals SDI1, SDI2, and SDI3 interface to the Left and Right Front Channels, Left and Right Rear Channels and the Center and LFE Channels respectively. The SDATA_OUT line outputs data from the two ADCs.

I2S LINE	ANALOG CHANNEL	FUNCTION
SDI1	Left & Right Front	I²S data interface to DAC_LF & DAC_RF
SDI2	Left & Right Rear	I²S data interface to DAC_LR & DAC_RR
SDI3	Center & LFE	I²S data interface to DAC_CTR & DAC_LFE
Alternate SDI1	All Channels	One line mode for all six channels
SDATA_OUT	Mic & Diff	I²S data interface from the ADCs

Table 1. Digital Audio Interface Configuration

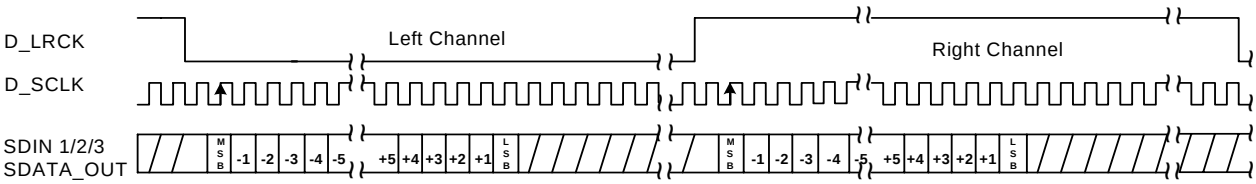

Figure 4. I²S Format

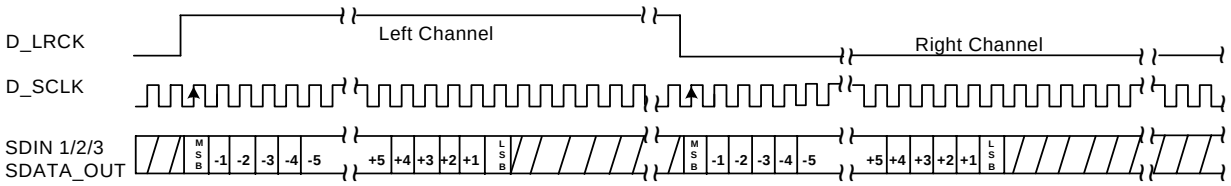

Figure 5. I²S Left Justified Format

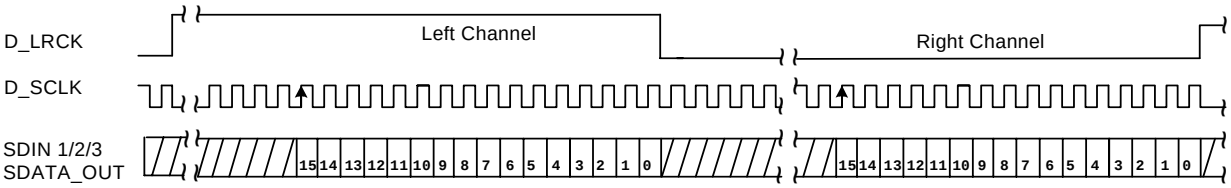

Figure 6. I²S Right Justified 16 Bit Format

6.2. Single Line Format

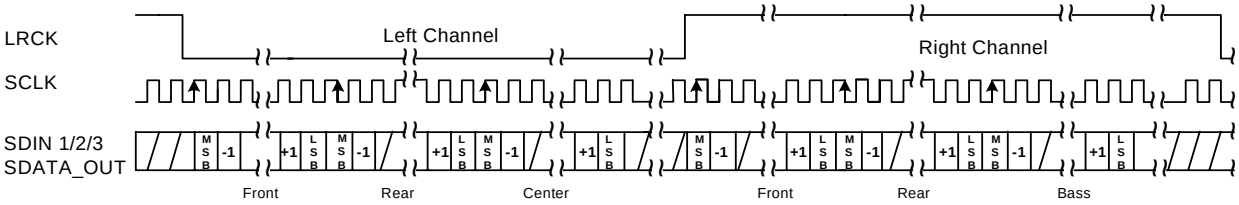

Figure 7. Single Line 20 Bit Data Mode Timing Diagram

Bits/Sample	SCLK Rate	Notes
20	128 Fs	6 inputs, 2 outputs, BRM only

Table 2. Single Line 20 Bit Data Mode, Data Valid on Rising Edge of SCLK

The Single line data mode for the STAC9460 allows data all six channels to be input to the chip on a single SDATA_IN line, SD1 (Pin 15) and output to all six analog outs (Pins 23-28). The ADC data will be valid during the first 20 SCLK cycles following a D_LRCLK transition.

6.3. I²C-Bus Interface

The I²C-Bus of the STAC9460 operates in compliance with the I²C-Bus Interface Specification from Philips Semiconductor. The I²C-Bus for the STAC9460 does include an Auto-Increment feature not identified by the Phillips specification. For Example, a typical write would have the following format: START....Chip Address (8 bits)....Register Address(8 bits)....Data(Register Address 8 bits)....Data (Register Address +1 8 bits)....Data (Register Address +2 8 bits)....STOP. The addresses will increment through the end of the address space or until a "STOP" condition (as per I²C spec) is received by the part. For detailed information relating to the I²C, please reference the I²C-Bus Interface Specification from Philips Semiconductor. Additional information for the Address Registers can be found in section 7.1.12

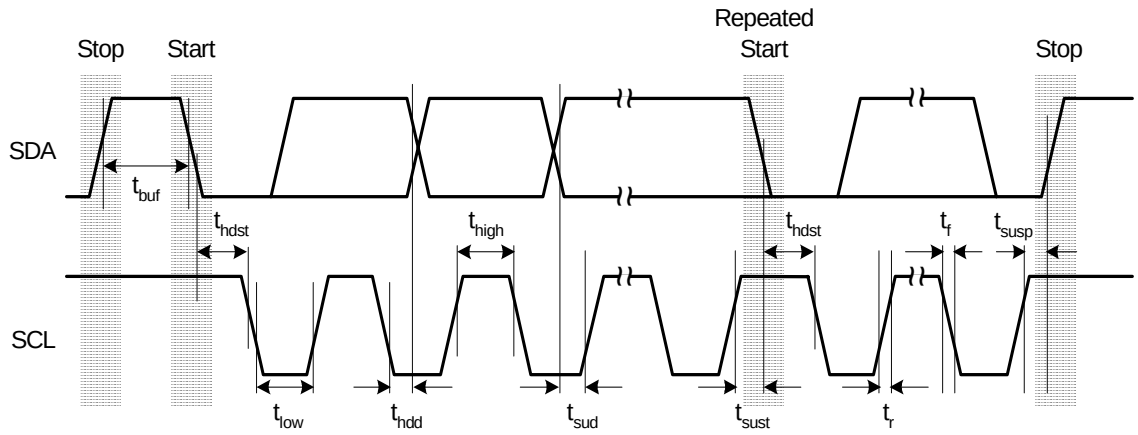

Figure 8. I²C Timing Diagram

PARAMETER	SYMBOL	MIN	MAX	UNITS
I^2C Mode (SDOUT < 47 kΩ to ground) (Note 1)				
SCL Clock Frequency	f_{scl}	-	100	KHz
Buss Free Time Between Transmissions	t_{buf}	4700	-	ns
Start Condition Hold Time (prior to first clock pulse)	t_{hdst}	4000	-	ns
Clock Low Time	t_{low}	4700	-	ns
Clock High Time	t_{high}	4000	-	ns
Setup Time for Repeated Start	t_{sust}	4700	-	ns
SDA Hold Time from SCL Falling (Note 2)	t_{hdd}	300	-	ns
SDA Setup Time to SCL Rising	t_{sud}	250	-	ns
Rise Time of Both SDA and SCL Lines	t_r	-	1000	ns
Fall Time of Both SDA and SCL Lines	t_f	-	300	ns
Setup Time for Stop Condition	t_{susp}	100	4700	ns

Note: 1. I^2C is a registered trademark of Philips Semiconductor and requires a license for use.
2. Data must be held for sufficient time to bridge the 300 ns transition time of SCL

Table 3. I^2C Mode Specifications

7. PROGRAMMABILITY

REG #	NAME	D7	D6	D5	D4	D3	D2	D1	D0	DEFAULT
00h	Reset/Status	RESERVED					REF	DIFF	MIC	00h
01h	Status	AR	AL	LFE	CTR	RR	LR	RF	LF	00h
02h	Master Volume	MMute	MV6	MV5	MV4	MV3	MV2	MV 1	MV0	80h
03h	LF Volume	Mute LF	LF6	LF5	LF4	LF3	LF2	LF1	LF0	80h
04h	RF Volume	Mute RF	RF6	RF5	RF4	RF3	RF2	RF1	RF0	80h
05h	LR Volume	Mute LR	LR6	LR5	LR4	LR3	LR2	LR1	LR0	80h
06h	RR Volume	Mute RR	RR6	RR5	RR4	RR3	RR2	RR1	RR0	80h
07h	Center Volume	Mute C	C6	C5	C4	C3	C2	C1	C0	80h
08h	LFE Volume	Mute LFE	B6	B5	B4	B3	B2	B1	B0	80h
09h	Mic L Volume	Mute ADCL	RESERVED			ML3	ML2	ML1	ML0	80h
0Ah	Mic R Volume	Mute ADCR	RESERVED			MR3	MR2	MR1	MR0	80h
0Bh	RESERVED									
0Ch	De-Emphasis	DEM1	DEM0	DEM LFE	DEM CTR	DEMRR	DEMLR	DEMRF	DEMLF	00h
0Dh	General Purpose	MD	HPFD	RSVD	HPFF	RESERVED				00h
0Eh	Audio Port Control	MSS	RESERVED		ADF4	ADF3	ADF2	ADF1	ADF0	00h
0Fh	Master Clocking	RESERVED			MCM2	MCM1	MCM0	SRM1	SRM0	10h
10h	Powerdown Ctrl	RESERVED				ADC Bias	VREF	DIG	DIFF	00h
11h	Powerdown Ctrl	ADR	ADL	PLFE	PCTR	PRR	PLR	PRF	PLF	00h
12h	Revision Code	0	0	0	0	0	0	0	0	00h
13h	Address Control Register	0	0	0	0	0	0	0	0	00h
14h	Address Register	0	1	0	1	0	1	CS	R/W	56h

Table 4. Programming Registers

1. All registers not shown are reserved.
2. Any bits marked “Reserved” should be written zero for normal operation

7.1. List of Registers

7.1.1. Reset/Status Register (00h)

Writing any value to this register performs a register reset, which causes all registers to revert to their default values. Reading this register returns the corresponding status of the chip section represented by the bits. The bits are defined below:

BIT	NAME	DESCRIPTION
D0	MIC	Microphone inputs
D1	DIFF	Differential inputs
D2	REF	Reference
D3-D7	RESERVED	RESERVED

Table 5. Reset/Status Register

7.1.2. Status Register (01h)

Reading this (read only) register returns the corresponding status of the chip section represented by the bits. The bits are defined below:

BIT	NAME	DESCRIPTION
D0	LF	Left front channel
D1	RF	Right front channel
D2	LR	Left rear channel
D3	RR	Right rear channel
D4	CTR	Center channel
D5	LFE	Low Frequency Effects Channel
D6	AL	Left ADC
D7	AR	Right ADC

Table 6. Status Register

7.1.3. Master Volume Register (02h)

This register manages the output signal volume for all channels simultaneously and adds to the individual channel volume registers. The DAC range is from 0 to –96 dB with each step equivalent to approximately 0.75 dB. The MSB, bit D7, of the register is the mute bit for all DACs. When this bit is set, the output level is -∞ dB. Bits MV6.. MV0 are used to control the master volume.

MMUTE	MV6…MV0	FUNCTION
0	000 0000	0 dB Attenuation
0	111 1111	96 dB Attenuation
1	xxx xxxx	∞ dB Attenuation

Table 7. Master Volume Register

7.1.4. LF/RF, LR/RR, Center/LFE Output Channel Volume Registers(03h-08h)

These registers determines the output signal volumes ranging from 0 dB to –96 dB with 0.75 dB steps. The MSB of the register is the mute bit for the channel. When this bit is set to 1, the outputlevel for that channel is -∞ dB. Please note the STAC9462 only uses registers 3 and 4 for the Left and Right Front Channels.

D7	D6…D0	FUNCTION
0	000 0000	0 dB Attenuation
0	111 1111	96 dB Attenuation
1	XXX XXXX	∞ dB Attenuation

Table 8. DAC Digital Volume Registers

7.1.5. Microphone Input Volume Registers (09h-0Ah)

These registers control the gain/attenuation for each of the microphone inputs ranging from 0 dB to 10.5 dB, with each step corresponding to approximately 1.5dB.

MSB	ML3…ML0	FUNCTION
0	0000	0 dB
0	1111	22.5 dB
1	XXXX	∞ dB Attenuation

MSB	MR3…MR0	FUNCTION
0	0000	0 dB
0	1111	22.5 dB
1	XXXX	∞ dB Attenuation

Table 9. Left and Right Mic Input Volume Registers

7.1.6. De-Emphasis Register (0Ch)

This register is used to turn de-emphasis on and off for each channel. De-emphasis control bits D5..D0 (DEMLFE .. DEMLF) select whether or not de-emphasis is turned on or off for the given DAC and bits D7-D6 (DEM1 and DEM0) determine which response curve to use.

BIT	FUNCTION
D7	De-emphasis OFF
D6	De-emphasis ON
D5	De-emphasis select for the LFE
D4	De-emphasis select for the CENTER
D3	De-emphasis select fot the RIGHT REAR
D2	De-emphasis select for the LEFT REAR
D1	De-emphasis select for the RIGHT FRONT
D0	De-emphasis select for the LEFT FRONT

Table 10. On/Off De-emphasis Selection for Each Channel

D7 .. D6	FUNCTION
00	32 kHz Response Curve
01	44.1 kHz Response Curve
10	48 kHz Response Curve
11	96 kHz Response Curve

Table 11. De-emphasis Filter Selection

7.1.7. General Purpose Register (0Dh)

The MSB (MD) selects between the MIC inputs and the DIFF inputs being routed to the ADCs. Bit D6 (HPFD) disables the ADC pass filter capability. The ADC High Pass FIlter Freeze, bit D4, maintains the high pass filter output's current DC offset

HPFF	FUNCTION
0	ADC High Pass Filter Not-Frozen
1	ADC High Pass Filter Frozen

Table 12. Bit D4: HPFD

MD	FUNCTION
0	Differential Input Selected
1	Microphone Selected

Table 13. Bit D5 Microphone Differential Mux By-Pass Control

HPFD	FUNCTION
0	ADC High Pass Filter Enabled
1	ADC High Pass Filter Disabled

Table 14. Bit D6 ADC High Pass Filter Disable

7.1.8. Audio Port Control (0Eh)

The I^2S port is controlled via the bits contained in this register. Formatting is controlled by the Audio Data Format bits, ADF4..ADF0 (bits D4 .. D0) of the register. The audio formats available are standard I^2S, Left Justified, Right Justified (16, 20, or 24-Bit) and One Line.

ADF4 .. ADF0 D4..D0	AUDIO DATA FORMAT
00000	I^2S
00001	Left Justified
00010	16 bit Right Justified
01010	20 bit Right Justified
10010	24 bit Right Justified
00011	One Line
*	All settings not shown are Reserved

Table 15. Audio Data Format Selection

7.1.9. *Master Clocking Register (0Fh)*

The Master Clocking Register is used to set the sampling rate of the converters to one of the three sample rate modes. Base rate, mid rate and high rate are selected with bits D1-D0 of the register. The master clock mode is set by selecting bits D4,,D2. The master clock mode is used generate an internal clock of the correct frequency based on the MCLK supplied by the user. For example, in the default mode (MCM=100 and SRM=00), MCLK in 512x, so if the sample rate is 48kHz then MCLK must be at 24.576MHz, which is 512x48kHz. Or with MCM=011 and SRM=01 (Mid Rate Mode) and a sample rate of 96kHz, then MCLK must be at 192x96kHz which is 18.432MHz. THe Master CLocking Register should be set before unmuting any DAC or ADC channels in register 02h to 0Ah. MCM2 (D4) set is default mode.

SRM1SRM0 D1,D0	SAMPLE RATE	FUNCTION
00 (default)	SR ≤ 48 kHz	Base Rate Mode
01	48 kHz < SR ≤ 96 kHz	Mid Rate Mode
10	96 kHz < SR ≤ 192kHz	High Rate Mode*
11	Reserved	Reserved

Note:*ADC operates in Mid Rate Mode when High Rate Mode is selected, and will send each sample for two successive I2S frames.

Table 16. Sample Rate Mode

MCM2 .. MCM0 D4,D3,D2	MCLK Mode*		
	BRM	MRM	HRM
000	128x	64x	32x
001	Reserved	Reserved	Reserved
010	256x	128x	64x
011	384x	192x	96x
100 (default)	512x	256x	128x
101	768x	384x	192x
110	Reserved	Reserved	Reserved
111	Reserved	Reserved	Reserved

*Note: MCLK rate is relative to sample rate. (MCM * SR = MCLK). The number of D_SCLKs/D_LRCLK is independant of the MCLK mode for the STAC9460/62, but most controllers will generate D_SCLK at 1/2, 1/4, 1/8, or 1/16 the MCLK rate.

Table 17. MCLK Mode

7.1.10. Powerdown Control Registers (10h-11h)

The **STAC9460** is capable of operating at reduced power when no activity is required. The state of power down is controlled by the Powerdown register. There are 11 separate power down commands. The Powerdown options are listed in Table 21. The bits can be used individually or in combination with each other, and control power distribution to the ADC's and DAC's. If the VREF option is selected it powers down the entire chip. Please note the rear, center, ad bass DAC's should be powered down for operation with the STAC9462.

REG 11H	
BIT	**FUNCTION**
D0: PLF	DAC_LF
D1: PRF	DAC_RF
D2: PLR	DAC_LR
D3: PRR	DAC_RR
D4: PCTR	DAC_CTR
D5: PLFE	DAC_LFE
D6: ADL	ADC Left
D7: ADR	ADC Right

REG 10H	
BIT	**FUNCTION**
D0: DIFF	Differential
D1: DIG	Powers down I^2S and De-emphasis
D2: VREF	Voltage Reference
D3: ADC Bias	Powerdown ADC Bias circuitry
D4:	X
D5:	X
D6:	X
D7:	X

* Note: Powers down I^2S and De-emphasis
X: denotes reserved

Table 18. Powerdown Control

7.1.11. Revision Code Register (12h)

The device Revision register contains a software readable revision-specific code used to identify performance, architectural, or software differences between various device revisions.

7.1.12. Address Control Register/Address Register (13h-14h)

The address for the chip is defaulted to 55/56 or 54/55 for read and write. The LSB (bit D1) is programmable with the CS, pin 10. The Address Register (14h) can be changed. To change the address, AB must first be written to The Address Control Register (13h). A soft or hard reset will reset the Address Register to its default value with CS representing the LSB (D1).

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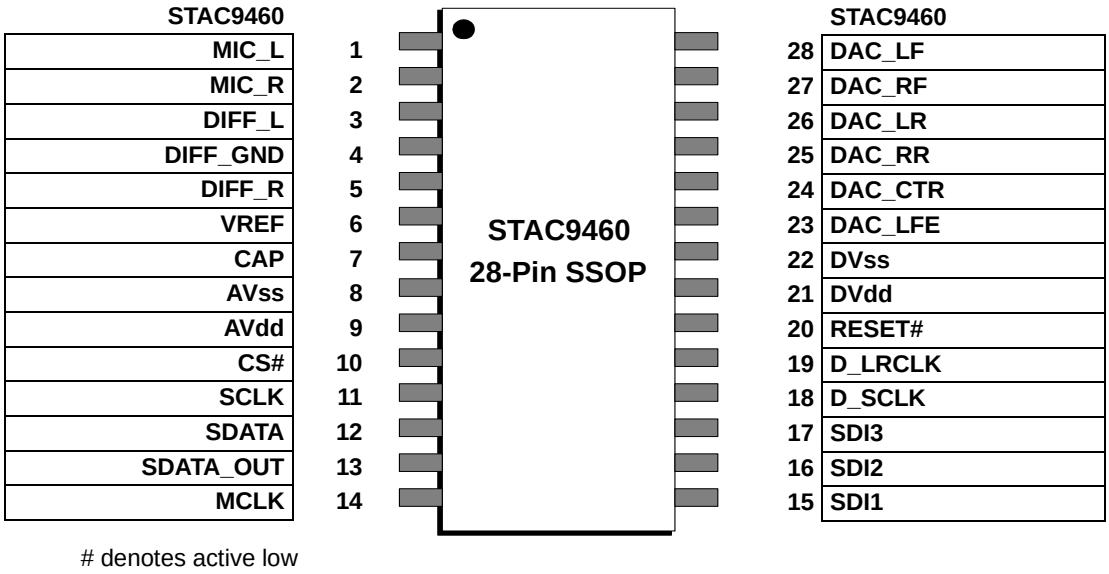
8. PIN DESCRIPTION

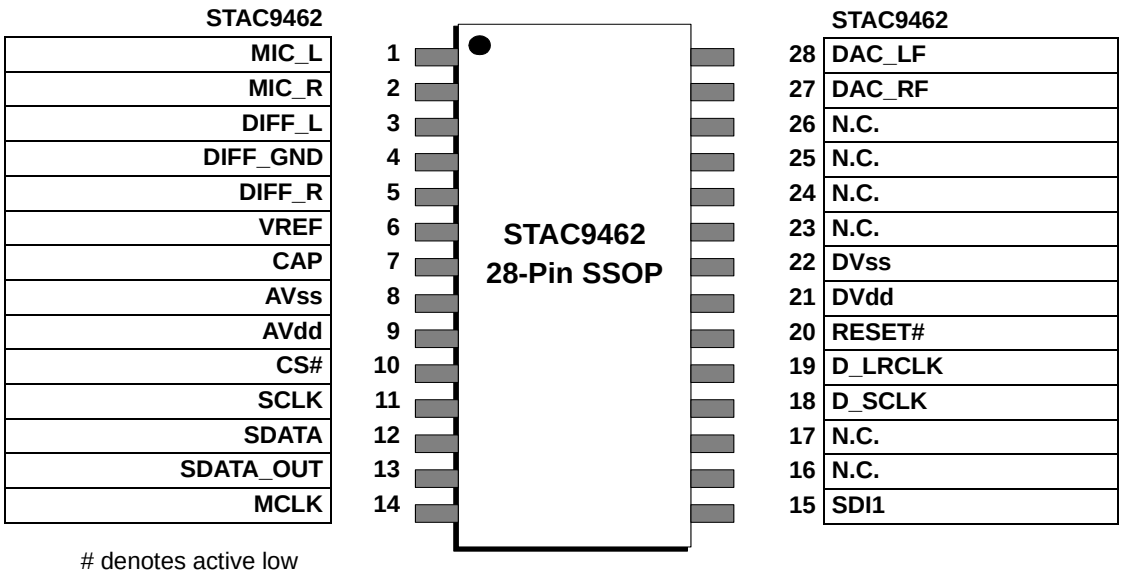
8.1. STAC9460 Pin and Signal Description



# denotes active low

Figure 9. STAC9460 Pin Designation

8.2. STAC9462 Pin and Signal Description



# denotes active low

Figure 10. STAC9462 Pin Designation

8.3. Digital I/O

These signals connect the **STAC9460/62** to an external µC/µP, DSP, and an external crystal.

SIGNAL NAME	TYPE	DESCRIPTION
RESET #	I	Master Hareware Reset
MCLK	I	See Table 17 on page 17 for details
CS#	I	Chip select
SCLK	I	Serial data clock
SDATA	I/O	Serial data input/output
SDI1	I	I^2S digital data input for LF and RF channels
SDI2	I	I^2S digital data input for LR and RR channels (STAC9460 only)
SDI3	I	I^2S digital data input for Center and LFEchannels (STAC9460 only)
SDATA_OUT	O	I^2S Serial data output for ADC outputs
D_LRCLK	I	I^2S digital data left/right clock
D_SCLK	I	I^2S digital data bit clock

# denotes active low

Table 19. Digital Signal List

8.4. Analog I/O

These signals connect the **STAC9460/62** to analog sources and sinks, including microphones and seakers.

SIGNAL NAME	TYPE	DESCRIPTION
MIC_L	I	Left microphone input
MIC_R	I	Right microphone input
DIFF_L	I	Left differential input
DIFF_GND	I	Differential input common ground
DIFF_R	I	Right differential input
DAC_LF	O	Left front channel (LF)
DAC_RF	O	Right front channel (RF)
DAC_LR	O	Left rear channel (LR) (STAC9460 only)
DAC_RR	O	Right rear channel (RR) (STAC9460 only)
DAC_CTR	O	Center channel (CTR) (STAC9460 only)
DAC_LFE	O	Low Frequency Effects output (LFE) (STAC9460 only)

Table 20. Analog Signal List

8.5. Filter/References

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SIGNAL NAME	TYPE	DESCRIPTION
VREF	O	Reference Voltage
CAP	O	ADC reference Cap

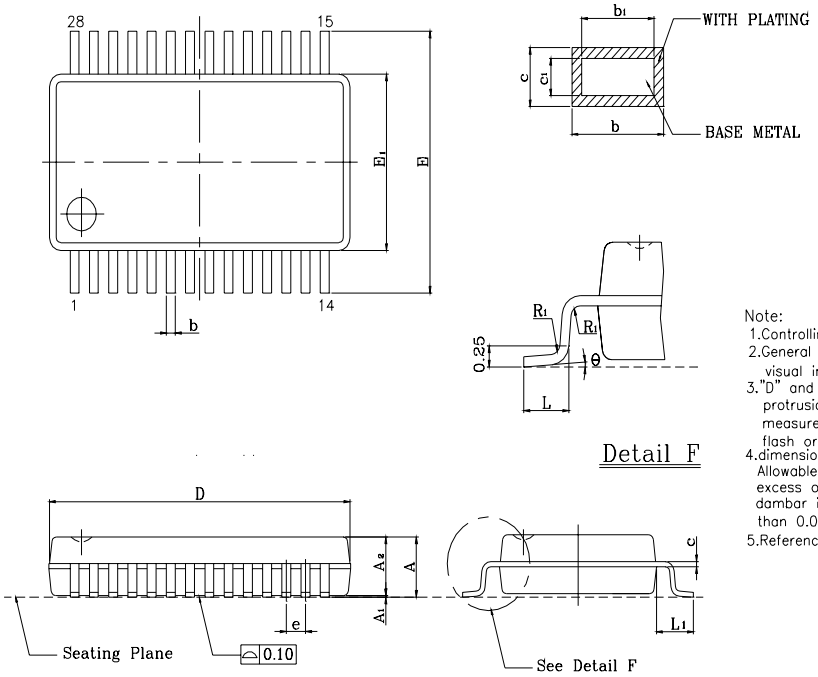
Table 21. Filtering and Voltage References

8.6. Power and Ground Signals

SIGNAL NAME	TYPE	DESCRIPTION
AVdd	I	Analog Vdd = 5.0 V
AVss	I	Analog Gnd
DVdd	I	Digital Vdd = 5.0 V or 3.3 V
DVss	I	Digital Gnd

Table 22. Power Signal List

9. PACKAGE DRAWING



Symbol	Dimension in inch			Dimension in mm		
	Min	Nom	Max	Min	Nom	Max
A	—	—	0.079	—	—	2.0
A_1	0.002	—	—	0.05	—	—
A_2	0.065	0.069	0.073	1.65	1.75	1.85
b	0.009	—	0.015	0.22	—	0.38
b_1	0.009	0.012	0.013	0.22	0.30	0.33
c	0.004	—	0.010	0.09	—	0.25
c_1	0.004	0.006	0.008	0.09	0.15	0.21
D	0.390	0.402	0.413	9.90	10.20	10.50
E	0.291	0.307	0.323	7.40	7.80	8.20
E_1	0.197	0.209	0.220	5.00	5.30	5.60
e	0.0256 BSC			0.65 BSC		
L	0.021	0.030	0.037	0.55	0.75	0.95
L_1	0.050 REF			1.25 REF		
R_1	0.004	—	—	0.09	—	—
θ	0°	4°	8°	0°	4°	8°

Note:
1. Controlling dimension: mm
2. General appearance spec should be based on final visual inspection spec.
3. "D" and "E1" dimensions do not include mold flash or protrusions but do include mold mismatch and are measured at datum plane -H- mold parting line. mold flash or protrusion shall not exceed 0.20 mm per side.
4. dimension b does not include dambar protrusion/intrusion Allowable dambar protrusion shall be 0.13mm total in excess of b dimension at maximum material condition. dambar intrusion shall not reduce dimension b by more than 0.07 mm at least material condition.
5. Reference document : JEDEC NO. MO-150AH