



# 4-Channel, Asynchronous Sample Rate Converter

# FEATURES

- AUTOMATIC SENSING OF INPUT-TO-OUTPUT SAMPLING RATIO
- WIDE INPUT-TO-OUTPUT SAMPLING RANGE: 16:1 to 1:16
- SUPPORTS INPUT AND OUTPUT SAMPLING RATES UP TO 212kHz
- DYNAMIC RANGE: 144dB (-60dbFS Input, BW = 20Hz to f<sub>s</sub>/2)
- THD+N: -140dB (0dbFS Input, BW = 20Hz to  $f_s/2$ )
- HIGH-PERFORMANCE, LINEAR-PHASE DIGITAL FILTERING WITH BETTER THAN 140dB OF STOP BAND ATTENUATION
- FLEXIBLE AUDIO SERIAL PORTS:
  - Master or Slave Mode Operation
  - Supports I<sup>2</sup>S, Left-Justified, Right-Justified, and TDM Data Formats
  - TDM Mode Allows Daisy-Chaining of Up to Four Devices
- SUPPORTS 24-, 20-, 18-, or 16-BIT INPUT AND OUTPUT DATA:
  - All Output Data is Dithered from the Internal 28-Bit Data Path
- SERIAL PERIPHERAL INTERFACE (SPI)<sup>™</sup> PORT SUPPORTS REGISTER READ AND WRITE OPERATIONS IN SOFTWARE MODE
- BYPASS MODE:
  Routes Input Port Data Directly to the Output Port
- FOUR GROUP DELAY OPTIONS FOR THE INTERPOLATION FILTER
- DIRECT DOWNSAMPLING OPTION FOR THE DECIMATION FILTER
- DIGITAL DE-EMPHASIS FILTER:
  User-Selectable for 32kHz, 44.1kHz, and 48kHz Sampling Rates
- SOFT MUTE FUNCTION
- PROGRAMMABLE DIGITAL OUTPUT ATTENUATION (SOFTWARE MODE ONLY):
   256 Steps: 0dB to -127.5dB with 0.5dB Steps

- INPUT-TO-OUTPUT SAMPLING RATIO READBACK (SOFTWARE MODE ONLY)
- POWER-DOWN MODES
- SUPPORTS OPERATION FROM A SINGLE +1.8V OR +3.3V POWER SUPPLY
- AVAILABLE IN A TQFP-64 PACKAGE

# **APPLICATIONS**

- DIGITAL MIXING CONSOLES
- DIGITAL AUDIO WORKSTATIONS
- AUDIO DISTRIBUTION SYSTEMS
- BROADCAST STUDIO EQUIPMENT
- GENERAL DIGITAL AUDIO PROCESSING

# DESCRIPTION

The SRC4194 is a four-channel, asynchronous sample rate converter (ASRC), designed for professional and broadcast audio applications. The SRC4194 combines a wide input-to-output sampling ratio with outstanding dynamic range and ultra low distortion. The input and output serial ports support the most common audio data formats, as well as a time division multiplexed (TDM) format. This allows the SRC4194 to interface to a wide range of audio data converters, digital audio receivers and transmitters, and digital signal processors.

The SRC4194 may be operated in Hardware mode as a standalone pin-programmed device, with dedicated control pins for serial port mode, audio data format, soft mute, bypass, and digital filtering functions. Alternatively, the SRC4194 may be operated in Software mode, where a four-wire serial peripheral interface (SPI) port provides access to internal control and status registers.

The SRC4194 operates from either a +1.8V core supply or a +3.3V core supply. When operating from +3.3V, the +1.8V required by the core logic is derived from an internal voltage regulator. The SRC4194 also requires a digital I/O supply, which operates from +1.65V to +3.6V. The SRC4194 is available in a TQFP-64 package.

U.S. Patent No. 7,262,716.



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

Core Supply Voltage
VDD18 –0.3V to +2.0V
VDD33 –0.3V to +4.0V
Digital I/O Supply Voltage, VIO –0.3V to +4.0V
Digital Input Voltage –0.3V to +4.0V
Operating Case Temperature Range, T <sub>C</sub> –40°C to +85°C
Storage Temperature Range, T <sub>STG</sub> 65°C to +150°C

(1) Stresses above these ratings may cause permanent damage. Exposure to absolute maximum conditions for extended periods may degrade device reliability. These are stress ratings only, and functional operation of the device at these or any other conditions beyond those specified is not supported.



This integrated circuit can be damaged by ESD. Texas Instruments recommends that all integrated circuits be handled with appropriate precautions. Failure to observe proper handling and installation procedures can cause damage.

ESD damage can range from subtle performance degradation to complete device failure. Precision integrated circuits may be more susceptible to damage because very small parametric changes could cause the device not to meet its published specifications.

#### **ORDERING INFORMATION**

For the most current package and ordering information, see the Package Option Addendum located at the end of this data sheet.

#### ELECTRICAL CHARACTERISTICS

All specifications at  $T_A = +25^{\circ}C$ , VDD33 = +3.3V,  $V_{IO} = +3.3V$ , REGEN = High, and VDD18 floating, unless otherwise noted.

			SRC4194		T
PARAMETER	CONDITIONS	MIN	TYP	MAX	UNITS
DYNAMIC PERFORMANCE					
Resolution			24		Bits
Input Sampling Frequency, f <sub>SIN</sub>		4		212	kHz
Output Sampling Frequency, f <sub>SOUT</sub>		4		212	kHz
INPUT/OUTPUT SAMPLING RATIO					
Upsampling				1:16	
Downsampling				16:1	
DYNAMIC RANGE <sup>(1)</sup>	BW = 20Hz to f <sub>SOUT</sub> /2, -60dBFS Input				
	f <sub>IN</sub> = 1kHz, A-Weighted				
44.1kHz:48kHz			143		dB
48kHz:44.1kHz			143		dB
48kHz:96kHz			143		dB
44.1kHz:192kHz			141		dB
96kHz:48kHz			144		dB
192kHz:12kHz			144		dB
192kHz:32kHz			144		dB
192kHz:48kHz			144		dB
32kHz:48kHz			143		dB
12kHz:192kHz			141		dB
TOTAL HARMONIC DISTORTION + NOISE(1)	BW = 20Hz to f <sub>SOUT</sub> /2, 0dBFS Input				
	f <sub>IN</sub> = 1kHz, Unweighted				
44.1kHz:48kHz			-140		dB
48kHz:44.1kHz			-140		dB
48kHz:96kHz			-140		dB
44.1kHz:192kHz			-137		dB
96kHz:48kHz			-140		dB
192kHz:12kHz			-140		dB
192kHz:32kHz			-141		dB
192kHz:48kHz			-141		dB
32kHz:48kHz			-140		dB
12kHz:192kHz			-137		dB
Interchannel Gain Mismatch			0		dB
Interchannel Phase Deviation			0		degrees

(1) Dynamic performance is measured with an Audio Precision System Two Cascade or Cascade Plus test system.

(2)  $f_{sMIN} = min (f_{sIN}, f_{sOUT}).$ 

(3)  $f_{sMAX} = max (f_{sIN}, f_{sOUT}).$ 

(4) Power-supply current for the power-down modes is measured without loading.

(5) Dynamic power-supply current measurements are performed with 2mA active loads on the excercized outputs.



				SRC4194	1	-	
PARAMETER		CONDITIONS	MIN	MIN TYP MAX		UNITS	
DIGITAL ATTENUATION		Software Mode Only					
Minimum				0		dB	
Maximum				-127.5		dB	
Step Size				0.5		dB	
Mute Attenuation		24-Bit Word Length		-144		dB	
DIGITAL INTERPOLATION FILTER CI	HARACTERI	STICS					
Passband	1				$0.4535  imes f_{SIN}$	Hz	
Passband Ripple	ĺ				±0.007	dB	
Transition Band			$0.4535 \times f_{SIN}$		$0.5465  imes f_{SIN}$	Hz	
Stop Band	i		$0.5465 \times f_{SIN}$			Hz	
Stop Band Attenuation			-144			dB	
Group Delay (64 sample buffer)	i	Decimation Filter Enabled		102.53125/f <sub>SIN</sub>		second	
Group Delay (64 sample buffer)	İ	Direct Downsampling Enabled		102/f <sub>sIN</sub>		second	
Group Delay (32 sample buffer)		Decimation Filter Enabled		70.53125/f <sub>SIN</sub>		second	
Group Delay (32 sample buffer)	İ	Direct Downsampling Enabled		70/f <sub>sIN</sub>		second	
Group Delay (16 sample buffer)		Decimation Filter Enabled		54.53125/f <sub>sIN</sub>		second	
Group Delay (16 sample buffer)		Direct Downsampling Enabled	ĺ	54/f <sub>sIN</sub>		second	
Group Delay (8 sample buffer)		Decimation Filter Enabled	ĺ	46.53125/f <sub>sIN</sub>		second	
Group Delay (8 sample buffer)		Direct Downsampling Enabled		46/f <sub>sIN</sub>		second	
DIGITAL DECIMATION FILTER CHAR	ACTERISTIC	S					
Passband	1				0.4535×f <sub>sOUT</sub>	Hz	
Passband Ripple					±0.008	dB	
Transition Band			0.4535×f <sub>s</sub> out		0.5465×f <sub>s</sub> out	Hz	
Stop Band			0.5465×f <sub>sOUT</sub>		0001	Hz	
Stop Band Attenuation			-143			dB	
Group Delay							
Decimation Filter		Decimation Filter Enabled		36.46875/f <sub>sOUT</sub>		second	
Direct Downsampling		Direct Downsampling Enabled		0		second	
DIGITAL DE-EMPHASIS							
De-Emphasis Error for f <sub>S</sub> = 32kHz, 44.1 48kHz	1kHz, or	De-Emphasis Enabled		0.001		dB	
DIGITAL I/O CHARACTERISTICS							
High-Level Input Voltage	VIH		$0.7 \times V_{IO}$		VIO	V	
Low-Level Input Voltage	VIL		0		$0.3 \times V_{IO}$	V	
High-Level Input Current	Чн			0.5	10	μA	
Low-Level Input Current	ι			0.5	10	μA	
High-Level Output Voltage	VOH	$I_{O} = -4mA$	0.8 × V <sub>IO</sub>		VIO	v	
Low-Level Output Voltage	VOL	$I_{O} = +4mA$	0		$0.2 \times V_{IO}$	V	
Input Capacitance	CIN			3		pF	
SWITCHING CHARACTERISTICS							
Reference Clock Timing							
RCKI Frequency(2)(3)			$128  imes f_{sMIN}$		50	MHz	
RCKI Period	<b>TROKIP</b>		20		1/(128×f <sub>sMIN</sub> )	ns	
RCKI Pulsewidth High	<sup>t</sup> RCKIH		0.4 × t <sub>RCKIP</sub>			ns	
RCKI Pulsewidth Low			0.4 × t <sub>RCKIP</sub>			ns	
Reset Timing			TONF				
RST Pulsewidth Low	<sup>t</sup> RSTL		500			ns	
Delay Following RST Rising Edge	NOTE	Software Mode Only	500			μs	
Input Serial Port Timing		,					
LRCKI to BCKI Setup Time	<sup>t</sup> LRIS		10			ns	
BCKI Pulsewidth High	tSIH		10			ns	
BCKI Pulsewidth Low	tSIL		10			ns	
	511		10	1	1	1 113	
SDIN Data Setup Time	tLDIS		10			ns	

(1) Dynamic performance is measured with an Audio Precision System Two Cascade or Cascade Plus test system.
 (2) f<sub>SMIN</sub> = min (f<sub>SIN</sub>, f<sub>SOUT</sub>).
 (3) f<sub>SMAX</sub> = max (f<sub>SIN</sub>, f<sub>SOUT</sub>).
 (4) Power-supply current for the power-down modes is measured without loading.
 (5) Dynamic power-supply current measurements are performed with 2mA active loads on the excercized outputs.



 $\label{eq:continued} \begin{array}{l} \textbf{ELECTRICAL CHARACTERISTICS (continued)} \\ \textbf{All specifications at } T_{A} = +25^{\circ}\text{C}, \ \text{VDD33} = +3.3\text{V}, \ \text{V}_{IO} = +3.3\text{V}, \ \text{REGEN} = \text{High, and } \text{VDD18 floating, unless otherwise noted.} \end{array}$ 

PARAMETER				SRC4194		
		CONDITIONS	MIN	ТҮР	MAX	UNITS
SWITCHING CHARACTERISTICS (cont	inued)					
Output Serial Port Timing						
SDOUT Data Delay Time	<sup>t</sup> DOPD				10	ns
SDOUT Data Hold Time	<sup>t</sup> DOH		2			ns
BCKO Pulsewidth High	<sup>t</sup> SOH		10			ns
BCKO Pulsewidth Low	<sup>t</sup> SOL		5			ns
TDM Mode Timing						
LRCKO Setup Time	<sup>t</sup> LROS		10			ns
LRCKO Hold Time	1_ROH		10			ns
TDMI Data Setup Time	tIDMS		10			ns
TDMI Data Hold Time	tDMH		10			ns
SPI Timing						
CCLK Frequency					25	MHz
CDATA Setup Time	<sup>t</sup> CDS		12			ns
CDATA Hold Time	<sup>t</sup> CDH		8			ns
CS Falling to CCLK Rising	<sup>t</sup> CSCR		15			ns
CCLK Falling to CS Rising	<sup>t</sup> CFCS		12			ns
CCLK Falling to CDOUT Data Valid	<sup>t</sup> CFDO				5	ns
CS Rising to CDOUT High Impedance	tCSZ				5	ns
POWER SUPPLIES(4, 5)	002					
Operating Voltage						
VDD18		REGEN = 0	+1.65	+1.8	+2.0	V
VDD33		REGEN = 1	+3.0	+3.3	+3.6	v
VIO		NEGEN = 1	+1.65	+3.3	+3.6	v
Supply Current		VDD18 = +1.8V, V <sub>IO</sub> = +1.8V, REGEN = 0	11.00	10.0	10.0	v
IDD, Hard Power-Down		$\overline{RST} = 0$ . No Clocks			100	μA
IDD, Soft Power-Down		$\overline{PDN}$ Bit = 0, No Clocks		100	100	μΑ
IDD, Dynamic		$f_{SIN} = 96$ kHz, $f_{SOUT} = 192$ kHz		80		mA
IIO, Hard Power-Down		$\frac{1}{\text{RST}} = 0$ . No Clocks		00	100	μA
IIO, Soft Power-Down		$\overline{PDN}$ Bit = 0, No Clocks		100	100	μΑ
IIO, Dynamic		$f_{SIN} = 96$ kHz, $f_{SOUT} = 192$ kHz		6		mA
Total Power Dissipation		$VDD18 = +1.8V, V_{IO} = +1.8V, REGEN = 0$		0		IIIA
P <sub>D</sub> , Hard Power-Down		RST = 0, No Clocks			1	mW
P <sub>D</sub> , Soft Power-Down		$\overline{PDN}$ Bit = 0, No Clocks		360	I	μW
<b>D</b> .		,		155		mW
P <sub>D</sub> , Dynamic Supply Current		f <sub>sIN</sub> = f <sub>sOUT</sub> = 192kHz VDD33 = +3.3V, V <sub>IO</sub> = +3.3V, REGEN = 1		155		IIIVV
IDD. Hard Power-Down		RST = 0. No Clocks			100	μA
				C	100	· ·
IDD, Soft Power-Down		$\overline{\text{PDN}} \text{ Bit} = 0, \text{ No Clocks}$		6 90		mA mA
IDD, Dynamic IIO, Hard Power-Down		f <sub>SIN</sub> = 96kHz, f <sub>SOUT</sub> = 192kHz RST = 0, No Clocks		90	100	
,		RST = 0, No Clocks PDN Bit = 0, No Clocks		100	100	μA
IIO, Soft Power-Down		·		100		μA
IIO, Dynamic		$f_{SIN} = 96$ kHz, $f_{SOUT} = 192$ kHz		6		mA
Total Power Dissipation		$VDD33 = +3.3V, V_{IO} = +3.3V, REGEN = 1$				
P <sub>D</sub> , Hard Power-Down		$\overline{RST} = 0$ , No Clocks			1	mW
P <sub>D</sub> , Soft Power-Down		PDN Bit = 0, No Clocks		21		mW
P <sub>D</sub> , Dynamic		f <sub>sIN</sub> = f <sub>sOUT</sub> = 192kHz		320		mW

(1) Dynamic performance is measured with an Audio Precision System Two Cascade or Cascade Plus test system.
 (2) f<sub>SMIN</sub> = min (f<sub>SIN</sub>, f<sub>SOUT</sub>).
 (3) f<sub>SMAX</sub> = max (f<sub>SIN</sub>, f<sub>SOUT</sub>).
 (4) Power-supply current for the power-down modes is measured without loading.
 (5) Dynamic power-supply current measurements are performed with 2mA active loads on the excercized outputs.



# PIN CONFIGURATION





### **PIN DESCRIPTIONS**

PIN #	NAME	I/O	DESCRIPTION
1	IFMTA0	Input	SRC A Audio Input Data Format <sup>(1)</sup>
2	IFMTA1	Input	SRC A Audio Input Data Format <sup>(1)</sup>
3	IFMTA2	Input	SRC A Audio Input Data Format <sup>(1)</sup>
4	OFMTA0	Input	SRC A Audio Output Data Format <sup>(1)</sup>
5	OFMTA1	Input	SRC A Audio Output Data Format <sup>(1)</sup>
6	OWLA0	Input	SRC A Audio Output Data Word Length <sup>(1)</sup>
7	OWLA1	Input	SRC A Audio Output Data Word Length <sup>(1)</sup>
8	BYPA	Input	SRC A Bypass Mode (Active High)
9	LGRPA0	Input	SRC A Low Group Delay Mode <sup>(1)</sup>
10	LGRPA1	Input	SRC A Low Group Delay Mode <sup>(1)</sup>
11	DDNA	Input	SRC A Direct Downsampling Mode (Active High) <sup>(1)</sup>
12	DEMA0	Input	SRC A Digital De-Emphasis Filter Mode <sup>(1)</sup>
13	DEMA1	Input	SRC A Digital De-Emphasis Filter Mode <sup>(1)</sup>
14	MODEA0	Input	SRC A Serial Port Mode <sup>(1)</sup>
15	MODEA1	Input	SRC A Serial Port Mode <sup>(1)</sup>
16	MODEA2	Input	SRC A Serial Port Mode <sup>(1)</sup>
17	RATIOA	Output	SRC A Ratio Flag
18	RDYA	Output	SRC A Ready Flag (Active Low)
19	MUTEA	Input	SRC A Output Soft Mute
20	RCKIA	Input	SRC A Reference Clock
21	RST	Input	Reset and Power-Down (Active Low)
22	H/S	Input	Control Mode (0 = Software, 1 = Hardware)
23	DGND	Ground	Digital Ground
24, 25	VDD33	Power	Core Supply, +3.3V. Required when REGEN is high. When REGEN is low, VDD33 must be left unconnected.
26	REGEN	Input	Voltage Regulator Enable (Active High)
27, 28	VDD18	Power	Core Supply, +1.8V. Required when REGEN is low. When REGEN is high, VDD18 must be left unconnected.
29	RCKIB	Input	SRC B Reference Clock
30	MUTEB	Input	SRC B Output Soft Mute
31	RDYB	Output	SRC B Ready Flag (Active Low)
32	RATIOB	Output	SRC B Ratio Flag
33	MODEB2 or CDIN	Input	SRC B Serial Port Mode <sup>(1)</sup> or SPI Port Serial Data Input <sup>(2)</sup> SRC B Serial Port Mode <sup>(1)</sup> or SPI Port Data Clock <sup>(2)</sup>
34	MODEB1 or CCLK	Input	
35	MODEB0 or CS	Input I/O	SRC B Serial Port Mode <sup>(1)</sup> or SPI Port Chip Select (Active Low) <sup>(2)</sup> SRC B Digital De-Emphasis Filter Mode <sup>(1)</sup> or SPI Port Serial Data Output <sup>(2)</sup>
36 37	DEMB1 or CDOUT DEMB0		SRC B Digital De-Emphasis Filter Mode(1) or SPI Port Serial Data Output(4)
38	DDNB	Input	SRC B Direct Downsampling Mode (Active High) <sup>(1)</sup>
38 39	LGRPB1	Input Input	SRC B Low Group Delay Mode <sup>(1)</sup>
39 40	LGRPB0	Input	SRC B Low Group Delay Mode(1)
40	BYPB	Input	SRC B Bypass Mode (Active High)
41	OWLB1	Input	SRC B Audio Output Data Word Length <sup>(1)</sup>
43	OWLB1	Input	SRC B Audio Output Data Word Length <sup>(1)</sup>
44	OFMTB1	Input	SRC B Audio Output Data Format <sup>(1)</sup>
45	OFMTB0	Input	SRC B Audio Output Data Format <sup>(1)</sup>
46	IFMTB2	Input	SRC B Audio Input Data Format <sup>(1)</sup>
47	IFMTB1	Input	SRC B Audio Input Data Format <sup>(1)</sup>
48	IFMTB0	Input	SRC B Audio Input Data Format <sup>(1)</sup>
49	SDOUTB	Output	SRC B Audio Output Data
50	BCKOB	I/O	SRC B Audio Output Bit Clock
51	LRCKOB	I/O	SRC B Audio Output Left/Right or Word Clock
52	TDMIB	Input	SRC B TDM Input Data (TDM Format Only)
53	BCKIB	I/O	SRC B Audio Input Bit Clock
54	LRCKIB	I/O	SRC B Audio Input Left/Right or Word Clock
55	SDINB	Input	SRC B Audio Input Data
56	VIO	Power	Digital I/O Supply, +1.65V to +3.6V
57	DGND	Ground	Digital Ground
58	SDINA	Input	SRC A Audio Input Data
59	LRCKIA	I/O	SRC A Audio Input Left/Right or Word Clock
60	BCKIA	I/O	SRC A Audio Input Bit Clock
00		1	
60 61	TDMIA	Input	SRC A TDM Input Data (TDM Format Only)
		Input I/O	SRC A TDM Input Data (TDM Format Only) SRC A Audio Output Left/Right or Word Clock
61	TDMIA		

(1) Disabled in Software control mode.
 (2) Disabled in Hardware control mode.

# TYPICAL CHARACTERISTICS

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# **TYPICAL CHARACTERISTICS (continued)**

All specifications at T<sub>A</sub> = +25°C, VDD33 = +3.3V, V<sub>IO</sub> = +3.3V, REGEN = High, and VDD18 floating, unless otherwise noted.



# **TYPICAL CHARACTERISTICS (continued)**

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# **TYPICAL CHARACTERISTICS (continued)**

All specifications at T<sub>A</sub> = +25°C, VDD33 = +3.3V, V<sub>IO</sub> = +3.3V, REGEN = High, and VDD18 floating, unless otherwise noted.



# **TYPICAL CHARACTERISTICS (continued)**

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# **TYPICAL CHARACTERISTICS (continued)**



# **TYPICAL CHARACTERISTICS (continued)**

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# **TYPICAL CHARACTERISTICS (continued)**



# **TYPICAL CHARACTERISTICS (continued)**

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# **TYPICAL CHARACTERISTICS (continued)**



# **TYPICAL CHARACTERISTICS (continued)**

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# **TYPICAL CHARACTERISTICS (continued)**





# **TYPICAL CHARACTERISTICS (continued)**





# **PRODUCT OVERVIEW**

The SRC4194 is a four-channel, asynchronous sample rate converter (ASRC), implemented as two stereo sections, referred to as SRC A and SRC B. Operation at input and output sampling frequencies up to 212kHz is supported, with a continuous input/output sampling ratio range of 16:1 to 1:16. Excellent dynamic range and THD+N are achieved by employing high-performance, linear-phase digital filtering with better than 140dB of image rejection. The digital filters provide settings for lower latency processing, including low group delay options for the interpolation filter and a direct downsampling option for the decimation filter. Digital de-emphasis filtering is included, supporting 32kHz, 44.1kHz, and 48kHz sampling frequencies.

The audio input and output ports support standard audio data formats, as well as a time division multiplexed (TDM) format. Word lengths of 24-, 20-, 18-, and 16-bits are supported. Input and output ports may operate in Slave mode, deriving their word and bit clocks from external input and output devices. Alternatively, one port may operate in Master mode while the other remains in Slave mode. In Master mode, the LRCK and BCK clocks are derived from the reference clock inputs, either RCKIA or RCKIB. The flexible configuration options for the input and output ports allow connections to a variety of audio data converters, digital audio interface devices, and digital signal processors.

A bypass mode is included, which allows audio data to be passed directly from the input port to the output port, bypassing the ASRC function. The bypass option is useful for passing through compressed or encoded audio data, as well as non-audio data (that is, control or status information).

A soft mute function is available for the SRC4194 in both Hardware and Software modes. Digital output attenuation is available only in Software mode. Both soft mute and digital attenuation functions provide artifact-free operation. The mute attenuation is typically -144dB, while the digital attenuation function is adjustable from 0dB to -127.5dB in 0.5dB steps.

The SRC4194 includes a four-wire SPI port, which is used to access on-chip control and status registers in Software mode. The SPI port facilitates interfacing to microprocessors or digital signal processors that support synchronous serial peripherals. In Hardware mode, dedicated control pins are provided for the majority of the SRC4194 functions. These pins can be hard-wired or driven by logic or host control.

#### FUNCTIONAL BLOCK DIAGRAM

Figure 1 shows a functional block diagram of the SRC4194. The SRC4194 is segmented into two stereo SRC sections, referred to as SRC A and SRC B. Each section can operate independently from the other. Each section has individual sets of configuration pins in Hardware mode, and separate banks of control and status registers in Software mode.



Figure 1. Functional Block Diagram



Operation for SRC A and SRC B is identical. Audio data is received at the input serial port, clocked by either the audio source device in Slave mode, or by the SRC4194 in Master mode. The output port data is clocked by either the audio output device in Slave mode, or by the SRC4194 in Master mode. The input data is passed through interpolation filters that upsample the data, which is then passed on to the re-sampler. The rate estimator compares the input and output sampling frequencies by comparing LRCKI, LRCKO, and a reference clock. The results of the rate estimation are utilized to configure the re-sampler coefficients and data pointers.

The output of the re-sampler is passed on to either the decimation filter or direct downsampler function. The decimation filter performs downsampling and anti-alias filtering functions, and is required when the output sampling frequency is equal to or lower than the input sampling frequency. The direct downsampler function does not provide any filtering, and may be used in cases when the output sampling frequency. The advantage of the direct downsampling function is a significant reduction in the group delay associated with the decimation function, allowing lower latency processing.

# **REFERENCE CLOCK**

The SRC4194 includes two reference clock inputs, one each for SRC A and SRC B. The reference clocks are applied at the RCKIA (pin 20) and RCKIB (pin 29) inputs, respectively. The reference clock is required for the rate estimator function, as well as for the input or output serial ports when configured in Master mode.

Figure 2 illustrates the reference clock connections and requirements for the SRC4194. When either the input or output port is configured in Master mode, the reference clock may operate at  $128f_s$ ,  $256f_s$ , or  $512f_s$ , where  $f_s$  is the desired sampling rate for the Master mode port. When both the input and output port are configured in Slave mode, the

reference clock does not have to be a multiple of the input or output sampling rates. The maximum reference clock input frequency is 50MHz for RCKIA and RCKIB.



Figure 2. Reference Clock Input Connections and Timing Requirements

### **RESET AND POWER-DOWN OPERATION**

The SRC4194 may be reset using the RST input (pin 21). There is no internal power on reset, so the user should force a reset sequence after power up in order to initialize the device. In order to force a reset, the reference clock inputs must be active, with external clock sources supplying a valid reference clock signal (refer to Figure 2). The user must assert RST low for a minimum of 500ns and then bring RST high again to force a reset. The reset function affects both SRC A and SRC B. Figure 3 shows the reset timing for the SRC4194.

In Software mode, there is a 500ms delay after the RST rising edge due to internal logic requirements. The customer should wait a minimum 500ms after the RST rising edge before attempting to write to the SPI port of the SRC4194 in Software mode.



Figure 3. Reset Pulsewidth Requirement

The SRC4194 also supports two power-down modes. The entire SRC4194 may be powered down by forcing and holding the RST input low. This is referred to as a Hard Power-Down, and the SRC4194 consumes the least amount of power in this mode.

In Software mode, there is an additional Soft Power-Down available, utilizing the PDN bit in Control Register 1. Soft Power-Down is enabled when the PDN bit is set to 0. Since SRC A and SRC B have their own separate register banks, they may be set to Soft Power-Down mode individually. During Soft Power-Down, the SPI port and control registers remain active for write and read access. The internal voltage regulator also remains active if the REGEN pin is forced high and +3.3V is applied at the VDD33 pin.

Soft Power-Down mode consumes more power than the Hard Power-Down mode. Refer to the Electrical Characteristics tables in this data sheet for supply current and power dissipation specifications for both modes.

Finally, there is one very important item to remember when using Software mode. The default state of the PDN bit is 0, meaning that the SRC4194 will default to the Soft Power-Down state for both SRC A and SRC B after power up or reset. The user must set the PDN bit to 1 for both the SRC A and SRC B control register banks in order to enable normal operation for both SRC sections.

### AUDIO SERIAL PORT MODES

The SRC4194 supports seven serial port modes for the SRC A and SRC B sections, which are shown in Table 1. In Hardware mode, the audio port mode is selected using the MODEA0 (pin 14), MODEA1 (pin 15), and MODEA2 (pin 16) inputs for SRC A, while the MODEB0 (pin 35), MODEB1 (pin 34), and MODEB2 (pin 33) inputs are used for SRC B.

In Software mode, the audio serial port modes are selected using the MODE[2:0] bits in Control Register 1 for the SRC A and SRC B register banks. The default setting for Software mode is both input and output ports set to Slave mode.

In Slave mode, the port LRCK and BCK clocks are configured as inputs, and receive their clocks from an external audio device. In Master mode, the LRCK and BCK clocks are configured as outputs, being derived from



the reference clock input for the corresponding SRC section, either RCKIA or RCKIB. Only one port can be set to Master mode at any given time, as indicated in Table 1.

Table 1.	Setting the	<b>Serial Port</b>	Modes (	x = A  or  B
----------	-------------	--------------------	---------	--------------

MODEx2	MODEx1	MODEx0	SERIAL PORT MODE
0	0	0	Both Input and Output Ports are Slave mode
0	0	1	Output Port is Master Mode with RCKIx = 128f <sub>S</sub>
0	1	0	Output Port is Master Mode with RCKIx = 512f <sub>S</sub>
0	1	1	Output Port is Master Mode with RCKIx = 256f <sub>S</sub>
1	0	0	Both Input and Output Ports are Slave mode
1	0	1	Input Port is Master Mode with RCKIx = 128fs
1	1	0	Input Port is Master Mode with RCKIx = 512f <sub>S</sub>
1	1	1	Input Port is Master Mode with RCKIx = 256f <sub>S</sub>

### **INPUT PORT OPERATION**

The audio input port is a three-wire synchronous serial interface that may operate in either Slave or Master mode. The SDINA (pin 58) and SDINB (pin 55) are the serial audio data inputs for SRC A and SRC B, respectively. Audio data is input at these pins in one of three standard audio data formats: Philips  $I^2S$ , Left-Justified, or Right-Justified. The audio data word length may be up to 24-bits for  $I^2S$  and Left-Justified formats, while the Right-Justified format supports 16-, 18-, 20-, or 24-bit data. The audio data is always Binary Two's Complement with the MSB first. Refer to Figure 4 for the input data formats and Figure 5 for the critical timing parameters, which are also listed in the Electrical Characteristics table.

The bit clock is either an input or output at BCKIA (pin 60) and BCKIB (pin 53). In Slave mode, the bit clock is configured as an input pin, and may operate at rates from  $32f_s$  to  $128f_s$ ,with a minimum of one clock cycle per data bit. In Master mode, bit clock operates at a fixed rate of  $64f_s$ .

The left/right word clock, LRCKIA (pin 59) and LRCKIB (pin 54), may be configured as an input or output pin. In Slave mode, left/right clock is an input pin, while in Master mode the left/right clock is an output pin. In either case, the clock rate is equal to  $f_s$ , the input sampling frequency. The LRCKI duty cycle is fixed to 50% for Master mode operation.





Figure 4. Input Data Formats



Figure 5. Input Port Timing

Table 2 illustrates the data format selection for the input port. For Hardware mode, the IFMTA0 (pin 1), IFMTA1 (pin 2), and IFMTA2 (pin 3) inputs are utilized to set the

input port data format for SRC A. IFMTB0 (pin 48), IFMTB1 (pin 47), and IFMTB2 (pin 46) are utilized to set the input port data format for SRC B.

Table 2. Input	Port Data	Format	Selection	(x = A  or  B)
----------------	-----------	--------	-----------	----------------

IFMTx2	IFMTx1	IFMTx0	INPUT PORT DATA FORMAT
0	0	0	24-Bit Left-Justified
0	0	1	24-Bit I <sup>2</sup> S
0	1	0	Unused
0	1	1	Unused
1	0	0	16-Bit Right-Justified
1	0	1	18-Bit Right-Justified
1	1	0	20-Bit Right-Justified
1	1	1	24-Bit Right-Justified

In Software mode, the IFMT[2:0] bits in Control Register 3 are used to select the data format for the SRC A and SRC B register banks. The default format is 24-bit Left-Justified.



#### OUTPUT PORT OPERATION

The audio output port is a four-wire synchronous serial interface that may operate in either Slave or Master mode. SDOUTA (pin 64) and SDOUTB (pin 49) are the serial audio data outputs for SRC A and SRC B, respectively. Audio data is output at these pins in one of four data formats: Philips I<sup>2</sup>S, Left-Justified, Right-Justified, or TDM. The audio data word length may be 16-, 18-, 20-, or 24-bits. For all word lengths, the data is triangular PDF dithered from the internal 28-bit data path. The data formats (with the exception of TDM mode) are shown in Figure 7, while critical timing parameters are shown in Figure 6 and listed in the Electrical Characteristics table. The TDM format and timing are shown in Figure 15 and Figure 16, respectively, while examples of standard TDM configurations are shown in Figure 17 and Figure 18.

The bit clock is either input or output at BCKOA (pin 63) and BCKOB (pin 50). In Slave mode, the bit clock is configured as an input pin, and may operate at rates from  $32f_s$  to  $128f_s$ , with a minimum of one clock cycle for each

data bit. The exception is the TDM mode, where the BCKO must operate at N  $\times$  64f<sub>s</sub>, where N is equal to the number of SRC sections cascaded on the TDM bus. In Master mode, the bit clock operates at a fixed rate of 64f<sub>s</sub> for all data formats except TDM, where BCKO operates at the reference clock frequency. Additional information regarding TDM mode operation is included in the Applications Information section of this data sheet.



Figure 6. Output Port Timing



Figure 7. Output Data Formats



The left/right word clock, LRCKOA (pin 62) and LRCKOB (pin 51), may be configured as an input or output pin. In Slave mode, the left/right clock is an input pin, while in Master mode it is an output pin. In either case, the clock rate is equal to  $f_s$ , the output sampling frequency. The clock duty cycle is fixed to 50% for I<sup>2</sup>S, Left-Justified, and Right-Justified formats in Master mode. The pulse width is fixed to 32-bit clock cycles for the TDM format in Master mode.

Table 3 illustrates data format selection for the output port. In Hardware mode, the OFMTA0 (pin 4), OFMTA1 (pin 5), OWLA0 (pin 6), and OWLA1 (pin 7) inputs are utilized to set the output port data format and word length for SRC A. The OFMTB0 (pin 45), OFMTB1 (pin 44), OWLB0 (pin 43), and OWLB1 (pin 42) inputs are utilized to set the output port data format and word length for SRC B.

#### Table 3. Output Port Data Format/Word Length Selection (x = A or B)

OFMTx1	OFMTx0	OUTPUT PORT DATA FORMAT
0	0	Left-Justified
0	1	1 <sup>2</sup> S
1	0	TDM
1	1	Right-Justified
OWLx1	OWLx0	OUTPUT PORT DATA WORD LENGTH
0	0	24 Bits
0	1	20 Bits
1	0	18 Bits
1	1	16 Bits

In Software mode, the OFMT[1:0] and OWL[1:0] bits in Control Register 3 are used to select the data format and word length for the SRC A and SRC B register banks. The default format is Left-Justified data with a default word length of 24-bits.

# **BYPASS MODE**

The SRC4194 includes a bypass function for both SRC A and SRC B, which routes the input port data directly to the output port, bypassing the sample rate conversion block. Bypass mode may be invoked by forcing BYPA (pin 8) or BYPB (pin 41) high in either Hardware or Software mode. In Software mode, the bypass function may also be accessed using the BYPASS bit in Control Register 1 for the SRC A and SRC B register banks. For normal SRC operation, the bypass pins and control bits should be set to 0.

No dithering is applied to the output data in Bypass mode, and the digital attenuation, de-emphasis, and soft mute functions are also unavailable. Bypass mode is useful for passing through compressed or encoded audio data, as well as non-audio data (that is, control or status information).

# INTERPOLATION FILTER GROUP DELAY OPTIONS

The SRC4194 provides four group delay options for the digital interpolation filter, as shown in Table 4. These options allow the user to tailor the group delay for a given application by selecting the number of input samples buffered prior to the re-sampling function.

Table 4. Low	Group Delay	Configuration
	(x = A  or  B)	

LGRPx1	LGRPx0	BUFFER SIZE
0	0	64 Samples
0	1	32 Samples
1	0	16 Samples
1	1	8 Samples

In Hardware mode, the LGRPA0 (pin 9) and LGRPA1 (pin 10) inputs are used to select the group delay for SRC A, while LGRPB0 (pin 40) and LGRPB1 (pin 39) inputs are used for SRC B.

In Software mode, the LGRP[1:0] bits in Control Register 2 are used for the SRC A and SRC B register banks. The 64 sample buffer option is selected by default in Software mode.

# DIRECT DOWNSAMPLING OPTION

The SRC4194 decimation function allows the selection of a direct downsampling option, as shown in Table 5. Unlike the decimation filter, the direct downsampler does not provide anti-alias filtering. This makes the direct downsampler suitable for applications where the output sample rate is higher than the input sample rate. The advantage of the direct downsampler is that there is no group delay associated with the decimation function.

Table 5. Decimation Function Configuration (x = A or B)

DDNx	DECIMATION FUNCTION
0	Decimation Filter Enabled
1	Direct Downsampler Enabled

In Hardware mode, the DDNA (pin 11) input is used to select the direct downsampler for SRC A, while the DDNB (pin 38) input is used for SRC B.

In Software mode, the DDN bit in Control Register 2 is used to select the direct downsampler for the SRC A and SRC B register banks. The decimation filter is selected by default, with direct downsampling disabled.

### DIGITAL DE-EMPHASIS FILTER

The SRC4194 includes digital de-emphasis filtering following the input serial ports. The de-emphasis filter processes audio data that has been pre-emphasized using the  $50/15\mu$ s transfer function, commonly used in consumer and professional audio systems. Pre-emphasis is utilized to increase the amplitude of the higher frequency components within the audio band. The de-emphasis filter normalizes the frequency response over the audio band.

The SRC4194 supports three sampling frequencies for the de-emphasis filter: 32kHz, 44.1kHz, and 48kHz. The de-emphasis filter can also be disabled. Table 6 shows the configuration table for the de-emphasis filter options.

#### Table 6. Digital De-Emphasis Filter Configuration (x = A or B)

DEMx1	DEMx0	DE-EMPHASIS FILTER FUNCTION
0	0	Disabled
0	1	48kHz Input Sample Rate
1	0	44.1kHz Input Sample Rate
1	1	32kHz Input Sample Rate

In Hardware mode, the DEMA0 (pin 12) and DEMA1 (pin 13) inputs are used to select the de-emphasis filter for SRC A, while DEMB0 (pin 37) and DEMB1 (pin 36) inputs are used for SRC B.

In Software mode, the DEM[1:0] bits in Control Register 2 are used to select the de-emphasis filter in both the SRC A and SRC B register banks. De-emphasis filtering is disabled by default in Software mode.

# SOFT MUTE FUNCTION

The soft mute function of the SRC4194 may be invoked by forcing the MUTEA (pin 19) or MUTEB (pin 30) inputs high. In Software mode, the mute function may also be accessed using the MUTE bit in Control Register 1 for either the SRC A and SRC B register banks. The soft mute function slowly attenuates the output signal level down to an all zeros output. For normal output, the soft mute function should be disabled by forcing the control pin or bit low. The soft mute function is disabled by default in Software mode.

#### DIGITAL ATTENUATION (Software Mode Only)

The SRC4194 includes independent digital attenuation for the Left and Right audio channels in Software mode. The attenuation ranges from 0dB (unity gain) to -127.5dB in 0.5dB steps. The attenuation settings are programmed using Control Register 4 and Control Register 5 for either the SRC A and SRC B register banks. The attenuation setting is programmed to 0dB (unity gain) by default.

Independent or Tracking attenuation modes. When TRACK = 0, the Left and Right channels are controlled independently. When TRACK = 1, the attenuation setting for the Left channel is also used for the Right channel, providing a tracking function. The digital attenuation mode is set to Independent by default.

# READY OUTPUT

The SRC4194 includes active low ready outputs for both SRC A and SRC B. The outputs are designated  $\overline{\text{RDYA}}$  (pin 18) and  $\overline{\text{RDYB}}$  (pin 31). The ready output is provided from the rate estimator block, with a low output state indicating that the input-to-output sampling frequency ratio has been determined and that the coefficients and address pointers for the re-sampling block have been updated. The ready signal may be used as a flag output for an external indicator or host.

The TRACK bit in Control Register 1 is used to select

# **RATIO OUTPUT**

The SRC4194 includes a sampling ratio flag output for both SRC A and SRC B. The outputs are designated RATIOA (pin 17) and RATIOB (pin 32). When the ratio output is low, it indicates that the output sampling frequency is lower than the input sampling frequency. When ratio output is high, it indicates that the output sampling frequency is higher than the input sampling frequency. The ratio output can be used as a flag output for either an external indicator or host.

#### SAMPLING RATIO READBACK (Software Mode Only)

In Software mode, Control Registers 6 and 7 in either the SRC A and SRC B register banks function as status registers, which contain the integer and fractional part of the input-to-output sampling ratio, or  $f_{SIN}$ : $f_{SOUT}$ . Given that  $f_{SOUT}$  or  $f_{SIN}$  is known, the unknown sampling rate can be computed using the contents of Registers 6 and 7. This function may be useful for controlling end application display or control processes. Refer to the *Control Register Definition* section of this datasheet for additional information regarding Registers 6 and 7.

#### SERIAL PERIPHERAL INTERFACE (SPI) PORT (Software Mode Only)

The SPI port is a four-wire synchronous serial interface used to access the on-chip control registers of the SRC4194. The interface is comprised of a serial data clock input, CCLK (pin 34); a serial data input, CDIN (pin 33); a serial data output, CDOUT (pin 36); and an active low chip-select input,  $\overline{CS}$  (pin 35). The CDOUT pin is a tri-state output and is forced to a high impedance state when the  $\overline{CS}$  input is forced high.







Figure 8 illustrates the protocol for register write and read operations via the SPI port. Figure 9 shows the critical timing parameters for the SPI port interface, which are listed in the Electrical Characteristics table.

Byte 0 indicates the register bank, register address, and read/write status for the operation. The functions contained within this byte are clearly shown in Figure 8. It should be noted that either one or both of the SRC A and SRC B register banks may be written to in the same operation, but only one bank can be selected at any time for a read operation. Byte 1 is a *don't care* byte. This byte is included in the protocol in order to maintain compatibility with current and future Texas Instruments' digital audio interface products, including the DIT4096, DIT4192, and SRC4193. Bytes 0 and 1 are followed by register data bytes.

As shown in Figure 8, a write or read operation starts by bringing the  $\overline{CS}$  input low. Bytes 0, 1, and 2 are then written to write or read a single register. Byte 2 is not needed for reading registers, so the CDIN pin can be forced low after Byte 0 for a read operation. Bringing the  $\overline{CS}$  input high after the third byte will write or read a single register address. However, if  $\overline{CS}$  remains low after writing or reading the first control or status byte, the port will automatically increment the address by 1, allowing successive addresses to be written or read sequentially. The address is automatically incremented by 1 after each byte is written or read, as long as the  $\overline{CS}$  input remains low. This is referred to as Auto-Increment operation, and is always enabled for the SPI port.









# CONTROL REGISTER MAP

(Software Mode Only)

The control register map for the SRC4194 is shown in Table 7. There are two identical register banks, one for SRC A and one for SRC B, each conforming to the register map shown in Table 7.

Register 0 is reserved for factory use and defaults to all zeros upon reset. The user should avoid writing to or reading this register, as unexpected operation may result if Register 0 is programmed to an arbitrary value.

Register 1 through Register 5 contain control bits, which are programmed to configure specific internal functions. Register 1 through Register 5 are available for write or read access. Register 6 and Register 7 contain the integer and fractional parts of the f<sub>sIN</sub>:f<sub>sOUT</sub> sampling ratio and are read only status registers.

		-	-			-		
REGISTER ADDRESS (HEX)	D7 (MSB)	D6	D5	D4	D3	D2	D1	D0
0	0	0	0	0	0	0	0	0
1	PDN	TRACK	0	MUTE	BYPASS	MODE2	MODE1	MODE0
2	0	0	0	DEM1	DEM0	DDN	LGRP1	LGRP0
3	OWL1	OWL0	OFMT1	OFMT0	0	IFMT2	IFMT1	IFMT0
4	AL7	AL6	AL5	AL4	AL3	AL2	AL1	AL0
5	AR7	AR6	AR5	AR4	AR3	AR2	AR1	AR0
6	SRI4	SRI3	SRI2	SRI1	SRI0	SRF10	SRF9	SRF8
7	SRF7	SRF6	SRF5	SRF4	SRF3	SRF2	SRF1	SRF0

Table 7. Control Register Map for Either the SRC A or SRC B Register Banks



#### CONTROL REGISTER DEFINITIONS (Software Mode Only)

This section contains descriptions for each control and status register available in Software mode. Reset defaults are also shown for each register bit.

#### **Register 1. System Control Register**

Ī	D7							D0
	(MSB)	D6	D5	D4	D3	D2	D1	(LSB)
	PDN	TRACK	0	MUTE	BYPASS	MODE2	MODE1	MODE0

#### MODE[2:0] Audio Serial Port Mode

These bits are used to select the Slave or Master mode status of the input and output serial ports.

MODE2	MODE1	MODE0	AUDIO SERIAL PORT MODE
0	0	0	Both Serial Ports are in Slave Mode (default)
0	0	1	Output Serial Port is Master with RCKI = 128f <sub>S</sub>
0	1	0	Output Serial Port is Master with RCKI = 512fs
0	1	1	Output Serial Port is Master with RCKI = 256f <sub>S</sub>
1	0	0	Both Serial Ports are in Slave Mode
1	0	1	Input Serial Port is Master with RCKI = 128f <sub>S</sub>
1	1	0	Input Serial Port is Master with RCKI = 512f <sub>S</sub>
1	1	1	Input Serial Port is Master with RCKI = 256f <sub>S</sub>

#### BYPASS Bypass Mode

This bit is logically OR'd with the bypass input (BYPA or BYPB) for the corresponding SRC section.

BYPASS	FUNCTION
0	Bypass Mode disabled with normal ASRC operation. (default)
1	Bypass Mode enabled with data routed directly from the input port to the output port, bypass- ing the ARSC function.

#### MUTE Output Soft Mute

This bit is logically OR'd with the MUTE input (MUTEA or MUTEB) for the corresponding SRC section.

MUTE	OUTPUT MUTE FUNCTION
0	Soft mute disabled. (default)
1	Soft mute enabled with output data attenuated to all 0s.

#### TRACK Digital Attenuation Tracking

TRACK	ATTENUATION TRACKING
0	Tracking Off: Attenuation for the Left and Right channels is controlled independently by Con- trol Register 4 and Control Register 5. (default)
1	Tracking On: Left channel attenuation setting is used for both channels.

#### PDN Power-Down

Setting this bit to 0 will force the corresponding SRC section into Soft Power-Down mode. All other register settings are preserved and the SPI port remains active. Setting this bit to 1 will power up the corresponding SRC section using the current register settings.

This bit defaults to 0 on power-up or reset. It must be programmed to 1 by the user in order to enable normal operation for the corresponding SRC section.



#### **Register 2. Digital Filter Control Register**

D7							D0
(MSB)	D6	D5	D4	D3	D2	D1	(LSB)
0	0	0	DEM1	DEM0	DDN	LGRP1	LGRP0

#### LGRP0 Interpolation Filter Group Delay

LGRP1 These bits are used to select the number of input samples to be stored in the data buffer before the re-sampler starts to process the data. This has a direct impact on the group delay or latency of the interpolation filter.

LGRP1	LGRP0	GROUP DELAY	
0	0	64 Samples (default)	
0	1	32 Samples	
1	0	16 Samples	
1	1	8 Samples	

#### DDN Decimation Filtering/Direct Downsampling

The DDN bit is used to enable or disable the direct downsampling function of the decimation block.

DDN	DECIMATION FILTER OPERATION
0	Decimation filter enabled. (default) (Must be used when $f_{\mbox{SOUT}}$ is less than or equal to $f_{\mbox{SIN}}$ .)
1	Direct downsampling enabled without filtering. (May be enabled when f <sub>SOUT</sub> is greater than f <sub>SIN</sub> .)

#### DEM0 Digital De-Emphasis Filtering

**DEM1** These bits are used to configure the digital de-emphasis filter function.

DEM1	DEM0	DE-EMPHASIS FILTER
0	0	Disabled (default)
0	1	48kHz Input Sampling Rate
1	0	44.1kHz Input Sampling Rate
1	1	32kHz Input Sampling Rate

#### Register 3. Audio Data Format Register

D7 (MSB)	D6	D5	D4	D3	D2	D1	D0 (LSB)
OWL1	OWL0	OFMT1	OFMT0	0	IFMT2	IFMT1	IFMT0

#### IFMT[2:0] Input Serial Port Data Format

These bits are utilized to select the audio data format for the input serial port.

IFMT2	IFMT1	IFMT0	INPUT FORMAT
0	0	0	24-Bit, Left-Justified (default)
0	0	1	24-Bit, I <sup>2</sup> S
0	1	0	Reserved
0	1	1	Reserved
1	0	0	Right-Justified, 16-Bit Data
1	0	1	Right-Justified, 18-Bit Data
1	1 0		Right-Justified, 20-Bit Data
1	1	1	Right-Justified, 24-Bit Data

#### OFMT[1:0] Output Port Data Format

These bits are utilized to select the audio data format for the output serial port.

OFMT1	OFMT0	OUTPUT FORMAT
0	0	Left-Justified (default)
0	1	I <sup>2</sup> S
1	0	TDM
1	1	Right-Justified

### OWL[1:0] Output Port Data Word Length

OWL1	OWL0	OUTPUT WORD LENGTH				
0	0	24-Bits (default)				
0	1	20-Bits				
1	0	18-Bits				
1	1	16-Bits				

#### Register 4. Digital Output Attenuation Register—Left Channel

D7 (MSB)	D6	D5	D4	D3	D2	D1	D0 (LSB)
AL7	AL6	AL5	AL4	AL3	AL2	AL1	AL0

This register is utilized to program the digital output attenuation for the Left output channel of the corresponding SRC section.

Register defaults to 00h, or 0dB (unity gain).

Output Attenuation (dB) =  $-N \times 0.5$ , where N = AL[7:0]<sub>DEC</sub>

#### Register 5. Digital Output Attenuation Register—Right Channel

	D7 (MSB)	D6	D5	D4	D3	D2	D1	D0 (LSB)
ľ	AR7	AR6	AR5	AR4	AR3	AR2	AR1	AR0

This register is utilized to program the digital output attenuation for the Right output channel of the corresponding SRC section. When the TRACK bit in Control Register 1 is set to 1, the Left Channel attenuation setting will also be used to set the Right Channel attenuation.

Register defaults to 00h, or 0dB (unity gain).

Output Attenuation (dB) =  $-N \times 0.5$ , where N = AR[7:0]<sub>DEC</sub>

#### Register 6. Sampling Ratio (read only)

D7							D0
(MSB)	D6	D5	D4	D3	D2	D1	(LSB)
SRI4	SRI3	SRI2	SRI1	SRI0	SRF10	SRF9	SRF8

#### Register 7. Sampling Ratio (read only)

D7 (MSB)	D6	D5	D4	D3	D2	D1	D0 (LSB)
SRF7	SRF6	SRF5	SRF4	SRF3	SRF2	SRF1	SRF0

The contents of Register 6 and Register 7 indicate the input-to-output sampling ratio, and can be used to determine either the input or output sampling rates when one of the two rates is known.

Bits SRI[4:0] comprise the integer portion of the input-to-output sampling ratio.

Bits SRF[10:0] comprise the fractional portion of the input-to-output sampling ratio.

The contents of Register 6 and Register 7 are updated when Register 6 is read. Register 6 must always be read first in order to obtain the latest ratio data for both registers.



# **APPLICATIONS INFORMATION**

This section provides practical applications information for hardware and systems engineers who will be designing the SRC4194 into their end equipment.

### **TYPICAL CONNECTIONS**

Figure 10 and Figure 11 illustrate typical connection diagrams for Hardware and Software modes, respectively. In Hardware mode, dedicated pins are controlled using external logic circuitry, hardwiring pins high or low, or by using the general-purpose I/O pins of a microprocessor or DSP. In Software mode, the SRC4194 is controlled via the 4-wire SPI port and optional GPIO from either a microprocessor or DSP.

Figure 12 illustrates the power-supply options for the SRC4194. When utilizing +3.3V for the core supply, the REGEN input (pin 26) must be driven high in order to enable the on-chip linear voltage regulator. The VDD33 pins are supplied with +3.3V and the VDD18 pins are left unconnected.

When utilizing +1.8V for the core supply, the REGEN input (pin 26) must be driven low in order to disable the on-chip linear voltage regulator. The VDD18 pins are supplied with +1.8V and the VDD33 pins are left unconnected.

Recommended power-supply bypass capacitor values are shown in Figure 10 through Figure 12. Ceramic capacitors (X7R chip type) are recommended for the 0.1 $\mu$ F capacitors, while the 10 $\mu$ F capacitors may be tantalum or multi-layer X7R ceramic chip type, or through-hole or surface-mount aluminum electrolytic capacitors.



Figure 10. Typical Pin Connections for Hardware Mode Operation











### INTERFACING TO DIGITAL AUDIO RECEIVERS AND TRANSMITTERS

The SRC4194 input and output ports are designed to interface to a variety of audio devices, including receivers and transmitters commonly used for AES/EBU, S/PDIF, and CP1201 communications. Texas Instruments manufactures the DIR1703 digital audio interface receiver and DIT4096/4192 digital audio transmitters to address these applications.

Figure 13 illustrates interfacing the DIR1703 to the SRC4194 input port. The DIR1703 operates from a single +3.3V supply, which requires that the  $V_{IO}$  supply (pin 56) for the SRC4194 to be set to +3.3V for interface compatibility.



Figure 13. Interfacing the SRC4194 to the DIR1703 Digital Audio Interface Receiver

Figure 14 shows the interface between the SRC4194 output port and the DIT4096 or DIT4192 audio serial port. Once again, the  $V_{IO}$  supplies for both the SRC4194 and DIT4096/4192 are set to +3.3V for interface compatibility.



# Figure 14. Interfacing the SRC4194 to the DIT4096/4192 Digital Audio Interface Receiver

Like the SRC4194 output ports, the DIT4096 and DIT4192 audio serial ports may be configured as a Master or Slave. In cases where the SRC4194 output port is set to Master mode and the DIT4096/4192 is configured as the Slave, it is recommended to use the reference clock source for the corresponding section of the SRC4194 as the master clock source for the DIT4096/4192. This will ensure that the transmitter audio serial port clocks, SYNC and SCLK, are synchronized to the master clock source.

### TDM APPLICATIONS

The SRC4194 supports a TDM output mode, which allows multiple devices to be daisy-chained together to create a serial frame. Each device occupies one sub-frame within a frame, and each sub-frame carries two channels (Left followed by Right). Each sub-frame is 64 bits long, with 32 bits allotted for each channel. The audio data for each channel is left-justified within the allotted 32 bits. Figure 16 illustrates the TDM frame format, while Figure 15 shows TDM input timing parameters, which are listed in the Electrical Characteristics table of this data sheet.



Figure 15. Input Timing for TDM Mode

The frame rate is equal to the output sampling frequency,  $f_s.$  The BCKO frequency for the TDM interface is  $N\times 64f_s,$  where N is the number of SRC sections included in the



daisy-chain. For Master mode, the output BCKO frequency is fixed to the reference clock input frequency. The number of devices that can be daisy-chained in TDM mode is dependent upon the output sampling frequency and the bit clock frequency, leading to the following numerical relationship.

Number of Daisy-Chained SRC Sections =  $(f_{BCKO}/f_s)/64$ 

Where:

f<sub>BCKO</sub> = Output Port Bit Clock (BCKO), 27MHz maximum

 $f_s$  = Output Port Sampling (or LRCKO) Frequency, 212kHz maximum.

This relationship holds true for both Slave and Master modes.

Figure 17 and Figure 18 show typical connection schemes for TDM mode. Although the TMS320C671x DSP family is shown as the audio processing engine in these figures, other TI digital signal processors with a multi-channel buffered serial port (McBSP<sup>™</sup>) may also function with this arrangement. Interfacing to processors from other manufacturers is also possible. Refer to the timing diagrams in this data sheet, along with the equivalent serial port timing diagrams shown in the DSP data sheet to determine compatibility.



Figure 16. TDM Frame Format





Figure 17. TDM Interface where All Devices are Slaves



Figure 18. TDM Interface where One Device is Master to Multiple Slaves

# **Revision History**

DATE	E REV PAGE SECTION		SECTION	DESCRIPTION
9/07	B 1 Front Page		Front Page	Added U.S. patent number.

NOTE: Page numbers for previous revisions may differ from page numbers in the current version.



### **PACKAGING INFORMATION**

Orderable part number	Status	Material type	Package   Pins	Package qty   Carrier	RoHS	Lead finish/	MSL rating/	Op temp (°C)	Part marking
	(1)	(2)			(3)	Ball material	Peak reflow		(6)
						(4)	(5)		
SRC4194IPAG	Active	Production	TQFP (PAG)   64	160   EIAJ TRAY (5+1)	Yes	NIPDAU	Level-4-260C-72 HR	-40 to 85	SRC4194I
SRC4194IPAG.A	Active	Production	TQFP (PAG)   64	160   EIAJ TRAY (5+1)	Yes	NIPDAU	Level-4-260C-72 HR	-40 to 85	SRC4194I
SRC4194IPAGR	Active	Production	TQFP (PAG)   64	1500   LARGE T&R	Yes	NIPDAU	Level-4-260C-72 HR	-40 to 85	SRC4194I
SRC4194IPAGR.A	Active	Production	TQFP (PAG)   64	1500   LARGE T&R	Yes	NIPDAU	Level-4-260C-72 HR	-40 to 85	SRC4194I
SRC4194IPAGT	Active	Production	TQFP (PAG)   64	250   SMALL T&R	Yes	NIPDAU	Level-4-260C-72 HR	-40 to 85	SRC4194I
SRC4194IPAGT.A	Active	Production	TQFP (PAG)   64	250   SMALL T&R	Yes	NIPDAU	Level-4-260C-72 HR	-40 to 85	SRC4194I

<sup>(1)</sup> **Status:** For more details on status, see our product life cycle.

<sup>(2)</sup> Material type: When designated, preproduction parts are prototypes/experimental devices, and are not yet approved or released for full production. Testing and final process, including without limitation quality assurance, reliability performance testing, and/or process qualification, may not yet be complete, and this item is subject to further changes or possible discontinuation. If available for ordering, purchases will be subject to an additional waiver at checkout, and are intended for early internal evaluation purposes only. These items are sold without warranties of any kind.

<sup>(3)</sup> RoHS values: Yes, No, RoHS Exempt. See the TI RoHS Statement for additional information and value definition.

<sup>(4)</sup> Lead finish/Ball material: Parts may have multiple material finish options. Finish options are separated by a vertical ruled line. Lead finish/Ball material values may wrap to two lines if the finish value exceeds the maximum column width.

<sup>(5)</sup> MSL rating/Peak reflow: The moisture sensitivity level ratings and peak solder (reflow) temperatures. In the event that a part has multiple moisture sensitivity ratings, only the lowest level per JEDEC standards is shown. Refer to the shipping label for the actual reflow temperature that will be used to mount the part to the printed circuit board.

<sup>(6)</sup> Part marking: There may be an additional marking, which relates to the logo, the lot trace code information, or the environmental category of the part.

Multiple part markings will be inside parentheses. Only one part marking contained in parentheses and separated by a "~" will appear on a part. If a line is indented then it is a continuation of the previous line and the two combined represent the entire part marking for that device.

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# PACKAGE OPTION ADDENDUM

23-May-2025



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# TAPE AND REEL INFORMATION





#### QUADRANT ASSIGNMENTS FOR PIN 1 ORIENTATION IN TAPE



*All dimensions are nominal												
Device	Package Type	Package Drawing		SPQ	Reel Diameter (mm)	Reel Width W1 (mm)	A0 (mm)	B0 (mm)	K0 (mm)	P1 (mm)	W (mm)	Pin1 Quadrant
SRC4194IPAGR	TQFP	PAG	64	1500	330.0	24.4	13.0	13.0	1.5	16.0	24.0	Q2
SRC4194IPAGT	TQFP	PAG	64	250	180.0	24.4	13.0	13.0	1.5	16.0	24.0	Q2



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# PACKAGE MATERIALS INFORMATION

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\*All dimensions are nominal

Device	Package Type	Package Drawing	Pins	SPQ	Length (mm)	Width (mm)	Height (mm)
SRC4194IPAGR	TQFP	PAG	64	1500	350.0	350.0	43.0
SRC4194IPAGT	TQFP	PAG	64	250	213.0	191.0	55.0

# TEXAS INSTRUMENTS

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### TRAY



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Chamfer on Tray corner indicates Pin 1 orientation of packed units.

*/	*All dimensions are nominal												
	Device	Package Name	Package Type	Pins	SPQ	Unit array matrix	Max temperature (°C)	L (mm)	W (mm)	K0 (µm)	P1 (mm)	CL (mm)	CW (mm)
Γ	SRC4194IPAG	PAG	TQFP	64	160	8 x 20	150	315	135.9	7620	15.2	13.1	13
Γ	SRC4194IPAG.A	PAG	TQFP	64	160	8 x 20	150	315	135.9	7620	15.2	13.1	13

# **MECHANICAL DATA**

MTQF006A - JANUARY 1995 - REVISED DECEMBER 1996

#### 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



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