

# 14-Bit, 160 MSPS TxDAC+<sup>®</sup> with 2× Interpolation Filter

# Preliminary Technical Data 6-14-00

AD9772A

## FEATURES

Single 2.8 V to 3.2 V Supply  
14-Bit DAC Resolution and Input Data Width  
160 MSPS Input Data Rate  
67.5 MHz Reconstruction Passband @ 160 MSPS  
75 dBc SFDR @ 25 MHz  
2× Interpolation Filter with High or Low Pass Response  
73 dB Image Rejection with 0.005 dB Passband Ripple  
“Zero-Stuffing” Option for Enhanced Direct IF Performance  
Internal 2×/4× Clock Multiplier  
<250 mW Power Dissipation; 13 mW with Power-Down Mode  
48-Lead LQFP Package

## APPLICATIONS

Communication Transmit Channel  
W-CDMA Base Stations, Multicarrier Base Stations,  
Direct IF Synthesis, Wideband Cable Systems  
Instrumentation

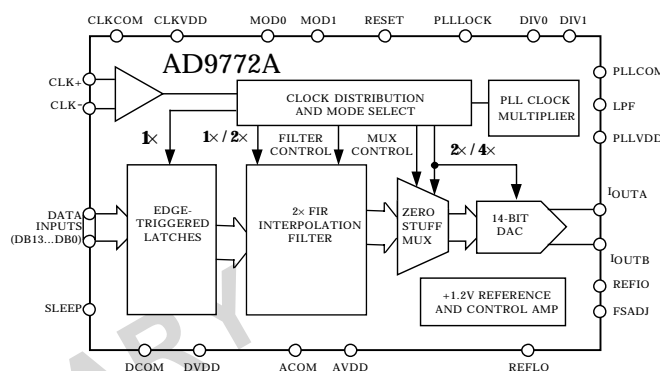
## PRODUCT DESCRIPTION

The AD9772A is a single supply, oversampling, 14-bit digital-to-analog converter (DAC) optimized for baseband or IF waveform reconstruction applications requiring exceptional dynamic range. Manufactured on an advanced CMOS process, it integrates a complete, low distortion 14-bit DAC with a  $2\times$  digital interpolation filter and clock multiplier. The on-chip PLL clock multiplier provides all the necessary clocks for the digital filter and the 14-bit DAC. A flexible differential clock input allows for a single-ended or differential clock driver for optimum jitter performance.

For baseband applications, the  $2\times$  digital interpolation filter provides a low pass response, hence providing up to a three-fold reduction in the complexity of the analog reconstruction filter. It does so by multiplying the input data rate by a factor of two while simultaneously suppressing the original upper inband image by more than 73 dB. For direct IF applications, the  $2\times$  digital interpolation filter response can be reconfigured to select the upper inband image (i.e., high pass response) while suppressing the original baseband image. To increase the signal level of the higher IF images and their passband flatness in direct IF applications, the AD9772A also features a “zero stuffing” option in which the data following the  $2\times$  interpolation filter is upsampled by a factor of two by inserting midscale data samples.

The AD9772A can reconstruct full-scale waveforms with bandwidths as high as 67.5 MHz while operating at an input data rate of 160 MSPS. The 14-bit DAC provides differential current outputs to support differential or single-ended applications. A segmented current source architecture is combined with a proprietary switching technique to reduce spurious components and enhance

### FUNCTIONAL BLOCK DIAGRAM



dynamic performance. Matching between the two current outputs ensures enhanced dynamic performance in a differential output configuration. The differential current outputs may be fed into a transformer or a differential op amp topology to obtain a single-ended output voltage using an appropriate resistive load.

The on-chip bandgap reference and control amplifier are configured for maximum accuracy and flexibility. The AD9772A can be driven by the on-chip reference or by a variety of external reference voltages. The full-scale current of the AD9772A can be adjusted over a 2 mA to 20 mA range, thus providing additional gain ranging capabilities.

The AD9772A is available in a 48-lead LQFP package and specified for operation over the industrial temperature range of  $-40^{\circ}\text{C}$  to  $+85^{\circ}\text{C}$ .

## PRODUCT HIGHLIGHTS

1. A flexible, low power  $2\times$  interpolation filter supporting reconstruction bandwidths of up to 67.5 MHz can be configured for a low or high pass response with 73 dB of image rejection for traditional baseband or direct IF applications.
2. A “zero-stuffing” option enhances direct IF applications.
3. A low glitch, fast settling 14-bit DAC provides exceptional dynamic range for both baseband and direct IF waveform reconstruction applications.
4. The AD9772A digital interface, consisting of edge-triggered latches and a flexible differential or single-ended clock input, can support input data rates up to 160 MSPS.
5. On-chip PLL clock multiplier generates all of the internal high speed clocks required by the interpolation filter and DAC.
6. The current output(s) of the AD9772A can easily be configured for various single-ended or differential circuit topologies.

TxDAC+ is a registered trademark of Analog Devices, Inc.

REV. PrA

Information furnished by Analog Devices is believed to be accurate and reliable. However, no responsibility is assumed by Analog Devices for its use, nor for any infringements of patents or other rights of third parties which may result from its use. No license is granted by implication or otherwise under any patent or patent rights of Analog Devices.

One Technology Way, P.O. Box 9106, Norwood, MA 02062-9106, U.S.A.  
Tel: 781/329-4700      World Wide Web Site: <http://www.analog.com>  
Fax: 781/326-8703      © Analog Devices, Inc., 1999

# AD9772A—SPECIFICATIONS

DC SPECIFICATIONS ( $T_{MIN}$  to  $T_{MAX}$ ,  $AVDD = +3.3$  V,  $CLKVDD = +3.3$  V,  $PLLVD = 0$  V,  $DVDD = +3.3$  V,  $I_{OUTFS} = 20$  mA, unless otherwise noted)

Parameter	Min	Typ	Max	Units
RESOLUTION	14			Bits
DC ACCURACY <sup>1</sup>				
Integral Linearity Error (INL)		±3.5		LSB
Differential Nonlinearity (DNL)		±2.0		LSB
Monotonicity (12-Bit)	Guaranteed Over Specified Temperature Range			
ANALOG OUTPUT				
Offset Error	−0.025		+0.025	% of FSR
Gain Error (Without Internal Reference)	−2	±0.5	+2	% of FSR
Gain Error (With Internal Reference)	−5	±1.5	+5	% of FSR
Full-Scale Output Current <sup>2</sup>		20		mA
Output Compliance Range	−1.0		+1.25	V
Output Resistance		200		kΩ
Output Capacitance		3		pF
REFERENCE OUTPUT				
Reference Voltage	1.14	1.20	1.26	V
Reference Output Current <sup>3</sup>		1		μA
REFERENCE INPUT				
Input Compliance Range	0.1		1.25	V
Reference Input Resistance (REFLO = 3 V)		10		MΩ
Small Signal Bandwidth		0.5		MHz
TEMPERATURE COEFFICIENTS				
Unipolar Offset Drift		0		ppm of FSR/°C
Gain Drift (Without Internal Reference)		±50		ppm of FSR/°C
Gain Drift (With Internal Reference)		±100		ppm of FSR/°C
Reference Voltage Drift		±50		ppm/°C
POWER SUPPLY				
AVDD				
Voltage Range	2.8	3.0	3.2	V
Analog Supply Current ( $I_{AVDD}$ )		34	37	mA
Analog Supply Current in SLEEP Mode ( $I_{AVDD}$ )		4.3	6	mA
DVDD1, DVDD2				
Voltage Range	2.8	3.0	3.2	V
Digital Supply Current ( $I_{DVDD1} + I_{DVDD2}$ )		29	33	mA
CLKVDD, PLLVDD <sup>4</sup> (PLLVDD = 3.0 V)				
Voltage Range	2.8	3.0	3.2	V
Clock Supply Current ( $I_{CLKVDD} + I_{PLLVDD}$ )		25		mA
CLKVDD (PLLVDD = 0 V)				
Voltage Range	2.8	3.0	3.2	V
Clock Supply Current ( $I_{CLKVDD}$ )		5.5	7	mA
Nominal Power Dissipation <sup>5</sup>		250		mW
Power Supply Rejection Ratio (PSRR) <sup>6</sup> – AVDD	−0.6		+0.6	% of FSR/V
Power Supply Rejection Ratio (PSRR) <sup>6</sup> – DVDD	−0.025		+0.025	% of FSR/V
OPERATING RANGE	−40		+85	°C

## NOTES

<sup>1</sup>Measured at  $I_{OUTA}$  driving a virtual ground.

<sup>2</sup>Nominal full-scale current,  $I_{OUTFS}$ , is 32× the  $I_{REF}$  current.

<sup>3</sup>Use an external amplifier to drive any external load.

<sup>4</sup>Measured at  $f_{DATA} = 100$  MSPS and  $f_{OUT} = 1$  MHz, DIV1, DIV0 = 0V.

<sup>5</sup>Measured with PLL enabled at  $f_{DATA} = 50$  MSPS and  $f_{OUT} = 1$  MHz.

<sup>6</sup>Measured over a 2.8 V to 3.2 V range.

Specifications subject to change without notice.

## DYNAMIC SPECIFICATIONS

( $T_{MIN}$  to  $T_{MAX}$ ,  $AVDD = +3.3$  V,  $CLKVDD = +3.3$  V,  $DVDD = +3.3$  V,  $PLLVD = 0$  V,  $I_{OUTFS} = 20$  mA, Differential Transformer Coupled Output, 50  $\Omega$  Doubly Terminated, unless otherwise noted)

Parameter	Min	Typ	Max	Units
<b>DYNAMIC PERFORMANCE</b>				
Maximum DAC Output Update Rate ( $f_{DAC}$ )	400			MSPS
Output Settling Time ( $t_{ST}$ ) (to 0.025%)		11		ns
Output Propagation Delay <sup>1</sup> ( $t_{PD}$ )		17		ns
Output Rise Time (10% to 90%) <sup>2</sup>		0.8		ns
Output Fall Time (10% to 90%) <sup>2</sup>		0.8		ns
Output Noise ( $I_{OUTFS} = 20$ mA)		50		pA $\sqrt{Hz}$
<b>AC LINEARITY-BASEBAND MODE</b>				
Spurious-Free Dynamic Range (SFDR) to Nyquist ( $f_{OUT} = 0$ dBFS)				
$f_{DATA} = 65$ MSPS; $f_{OUT} = 1.01$ MHz		82		dBc
$f_{DATA} = 65$ MSPS; $f_{OUT} = 10.01$ MHz		79		dBc
$f_{DATA} = 65$ MSPS; $f_{OUT} = 26.01$ MHz		74		dBc
$f_{DATA} = 160$ MSPS; $f_{OUT} = 2.02$ MHz		82		dBc
$f_{DATA} = 160$ MSPS; $f_{OUT} = 20.02$ MHz		81		dBc
$f_{DATA} = 160$ MSPS; $f_{OUT} = 52.02$ MHz		73		dBc
Two-Tone Intermodulation (IMD) to Nyquist ( $f_{OUT1} = f_{OUT2} = -6$ dBFS)				
$f_{DATA} = 65$ MSPS; $f_{OUT1} = 5.01$ MHz; $f_{OUT2} = 6.01$ MHz		82		dBc
$f_{DATA} = 65$ MSPS; $f_{OUT1} = 15.01$ MHz; $f_{OUT2} = 17.51$ MHz		78		dBc
$f_{DATA} = 65$ MSPS; $f_{OUT1} = 24.1$ MHz; $f_{OUT2} = 26.2$ MHz		76		dBc
$f_{DATA} = 160$ MSPS; $f_{OUT1} = 10.02$ MHz; $f_{OUT2} = 12.02$ MHz		80		dBc
$f_{DATA} = 160$ MSPS; $f_{OUT1} = 30.02$ MHz; $f_{OUT2} = 35.02$ MHz		78		dBc
$f_{DATA} = 160$ MSPS; $f_{OUT1} = 48.2$ MHz; $f_{OUT2} = 52.4$ MHz		71		dBc
Total Harmonic Distortion (THD)				
$f_{DATA} = 50$ MSPS; $f_{OUT} = 1.0$ MHz; 0 dBFS		-78		dB
$f_{DATA} = 65$ MSPS; $f_{OUT} = 10.01$ MHz; 0 dBFS		-77		dB
Signal-to-Noise Ratio (SNR)				
$f_{DATA} = 65$ MSPS; $f_{OUT} = 16.26$ MHz; 0 dBFS		76		dB
$f_{DATA} = 100$ MSPS; $f_{OUT} = 25.1$ MHz; 0 dBFS		71		dB
Adjacent Channel Power Ratio (ACPR)				
WCDMA with 4.1 MHz BW, 5 MHz Channel Spacing				
IF = 16 MHz, $f_{DATA} = 65.536$ MSPS		78		dBc
IF = 32 MHz, $f_{DATA} = 131.072$ MSPS		68		dBc
Four-Tone Intermodulation				
15.6 MHz, 15.8 MHz, 16.2 MHz and 16.4 MHz at -12 dBFS		88		dBFS
$f_{DATA} = 65$ MSPS, Missing Center				
<b>AC LINEARITY-IF MODE</b>				
Four-Tone Intermodulation at IF = 70 MHz				
68.1 MHz, 69.3 MHz, 71.2 MHz and 72.0 MHz at -20 dBFS		77		dBFS
$f_{DATA} = 52$ MSPS, $f_{DAC} = 208$ MHz				

## NOTES

<sup>1</sup>Propagation delay is delay from CLK input to DAC update.

<sup>2</sup>Measured single-ended into 50  $\Omega$  load.

Specifications subject to change without notice.

# AD9772A—SPECIFICATIONS

## DIGITAL SPECIFICATIONS ( $T_{MIN}$ to $T_{MAX}$ , $AVDD = +3.3$ V, $CLKVDD = +3.3$ V, $PLLVDD = +0$ V, $DVDD = +3.3$ V, $I_{OUTFS} = 20$ mA, unless otherwise noted)

Parameter	Min	Typ	Max	Units
<b>DIGITAL INPUTS</b>				
Logic “1” Voltage	2.1	3		V
Logic “0” Voltage		0	0.9	V
Logic “1” Current <sup>1</sup>	-10		+10	$\mu$ A
Logic “0” Current	-10		+10	$\mu$ A
Input Capacitance		5		pF
<b>CLOCK INPUTS</b>				
Input Voltage Range	0		3	V
Common-Mode Voltage	0.75	1.5	2.25	V
Differential Voltage	0.5	1.5		V
<b>PLL CLOCK ENABLED—FIGURE 1a</b>				
Input Setup Time ( $t_S$ )	0.2			ns
Input Hold Time ( $t_H$ )	1.8			ns
Latch Pulsewidth ( $t_{LPW}$ )	1.5			ns
<b>PLL CLOCK DISABLED—FIGURE 1b</b>				
Input Setup Time ( $t_S$ )	-1.2			ns
Input Hold Time ( $t_H$ )	3.2			ns
Latch Pulsewidth ( $t_{LPW}$ )	1.5			ns
CLK/PLL LOCK Delay ( $t_{OD}$ )		TBD		ns

### NOTES

<sup>1</sup>MOD0, MOD1, DIV0, DIV1, SLEEP, RESET have typical input currents of 15  $\mu$ A.

Specifications subject to change without notice.

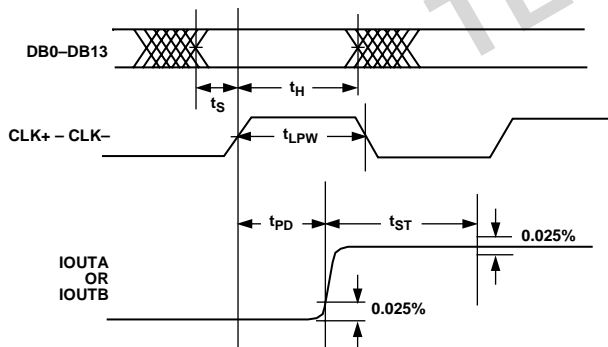


Figure 1a. Timing Diagram—PLL Clock Multiplier Enabled

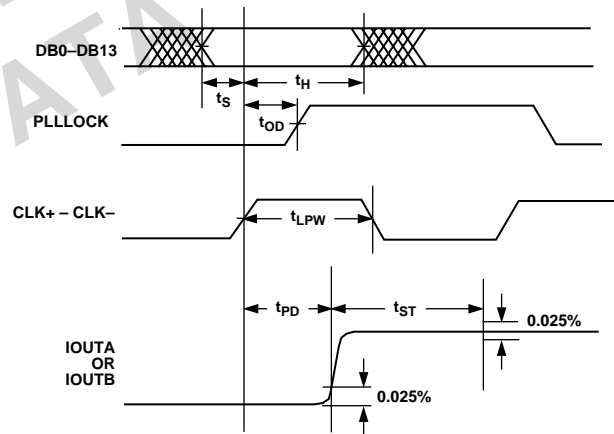


Figure 1b. Timing Diagram—PLL Clock Multiplier Disabled

# DIGITAL FILTER SPECIFICATIONS

( $T_{MIN}$  to  $T_{MAX}$ ,  $AVDD = +3.3$  V,  $CLKVDD = +3.3$  V,  $PLLVD = 0$  V,  $DVDD = +3.3$  V,  $I_{OUTFS} = 20$  mA, Differential Transformer Coupled Output,  $50\Omega$  Doubly Terminated, unless otherwise noted)

Parameter	Min	Typ	Max	Units
MAXIMUM INPUT DATA RATE ( $f_{DATA}$ )	150			MSPS
DIGITAL FILTER CHARACTERISTICS				
Passband Width <sup>1</sup> : 0.005 dB		0.401		$f_{OUT}/f_{DATA}$
Passband Width: 0.01 dB		0.404		$f_{OUT}/f_{DATA}$
Passband Width: 0.1 dB		0.422		$f_{OUT}/f_{DATA}$
Passband Width: -3 dB		0.479		$f_{OUT}/f_{DATA}$
LINEAR PHASE (FIR IMPLEMENTATION)				
STOPBAND REJECTION				
0.606 $f_{CLOCK}$ to 1.394 $f_{CLOCK}$		73		dB
GROUP DELAY <sup>2</sup>		21		Input Clocks
IMPULSE RESPONSE DURATION				
-40 dB		36		Input Clocks
-60 dB		42		Input Clocks

## NOTES

<sup>1</sup>Excludes  $\sin(x)/x$  characteristic of DAC.<sup>2</sup>Defined as the number of data clock cycles between impulse input and peak of output response.

Specifications subject to change without notice.

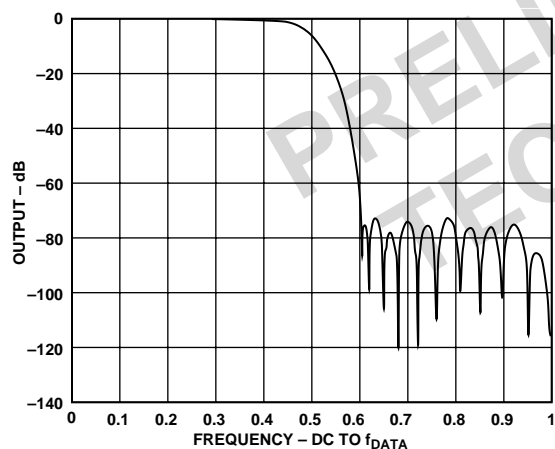


Figure 2a. FIR Filter Frequency Response—Baseband Mode

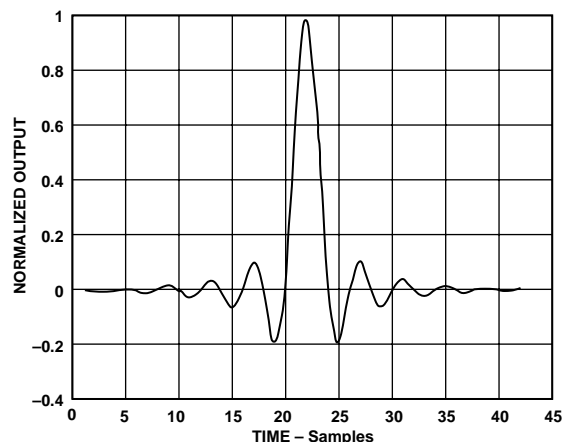


Figure 2b. FIR Filter Impulse Response—Baseband Mode

Table I. Integer Filter Coefficients for Interpolation Filter (43-Tap Half-Band FIR Filter)

Lower Coefficient	Upper Coefficient	Integer Value
H(1)	H(43)	10
H(2)	H(42)	0
H(3)	H(41)	-31
H(4)	H(40)	0
H(5)	H(39)	69
H(6)	H(38)	0
H(7)	H(37)	-138
H(8)	H(36)	0
H(9)	H(35)	248
H(10)	H(34)	0
H(11)	H(33)	-419
H(12)	H(32)	0
H(13)	H(31)	678
H(14)	H(30)	0
H(15)	H(29)	-1083
H(16)	H(28)	0
H(17)	H(27)	1776
H(18)	H(26)	0
H(19)	H(25)	-3282
H(20)	H(24)	0
H(21)	H(23)	10364
H(22)		16384

# AD9772A

## ABSOLUTE MAXIMUM RATINGS\*

Parameter	With Respect to	Min	Max	Units
AVDD, DVDD1-2, CLKVDD, PLLVDD	ACOM, DCOM, CLKCOM, PLLCOM	-0.3	+4.0	V
AVDD, DVDD1-2, CLKVDD, PLLVDD	AVDD, DVDD1-2, CLKVDD, PLLVDD	-4.0	+4.0	V
ACOM, DCOM1-2, CLKCOM, PLLCOM	ACOM, DCOM1-2, CLKCOM, PLLCOM	-0.3	+0.3	V
REFIO, REFLO, FSADJ, SLEEP	ACOM	-0.3	AVDD + 0.3	V
I <sub>OUTA</sub> , I <sub>OUTB</sub>	ACOM	-1.0	AVDD + 0.3	V
DB0-DB13, MOD0, MOD1, PLLLOCK	DCOM1-2	-0.3	DVDD + 0.3	V
CLK+, CLK-	CLKCOM	-0.3	CLKVDD + 0.3	V
DIV0, DIV1, RESET	CLKCOM	-0.3	CLKVDD + 0.3	V
LPF	PLLCOM	-0.3	PLLVDD + 0.3	V
Junction Temperature			+150	°C
Storage Temperature		-65	+150	°C
Lead Temperature (10 sec)			+300	°C

\*Stresses above those listed under Absolute Maximum Ratings may cause permanent damage to the device. This is a stress rating only; functional operation of the device at these or any other conditions above those indicated in the operational sections of this specification is not implied. Exposure to absolute maximum ratings for extended periods may effect device reliability.

## CAUTION

ESD (electrostatic discharge) sensitive device. Electrostatic charges as high as 4000 V readily accumulate on the human body and test equipment and can discharge without detection. Although the AD9772A features proprietary ESD protection circuitry, permanent damage may occur on devices subjected to high energy electrostatic discharges. Therefore, proper ESD precautions are recommended to avoid performance degradation or loss of functionality.



## ORDERING GUIDE

Model	Temperature Range	Package Description	Package Option*
AD9772AAST AD9772EB	-40°C to +85°C	48-Lead LQFP Evaluation Board	ST-48

\*ST = Thin Plastic Quad Flatpack.

## THERMAL CHARACTERISTIC

Thermal Resistance

48-Lead LQFP

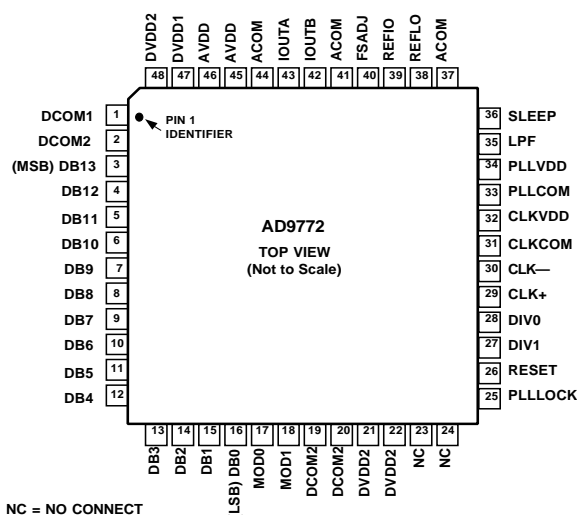
$$\theta_{JA} = 91^{\circ}\text{C/W}$$

$$\theta_{JC} = 28^{\circ}\text{C/W}$$

## PIN FUNCTION DESCRIPTIONS

Pin No.	Name	Description
1, 2, 19, 20	DCOM1, 2	Digital Common.
3	DB13	Most Significant Data Bit (MSB).
4–15	DB12–DB1	Data Bits 1–12.
16	DB0	Least Significant Data Bit (LSB).
17	MOD0	Invokes digital high-pass filter response (i.e., “half-wave” digital mixing mode). Active High.
18	MOD1	Invokes “zero-stuffing” mode. Active High. Note, “quarter-wave” digital mixing occurs with MOD0 also set HIGH.
23, 24	NC	No Connect, Leave Open.
21, 22, 47, 48	DVDD1, 2	Digital Supply Voltage (+2.8V to +3.2V).
25	PLLLOCK	Phase Lock Loop Lock Signal when PLL clock multiplier is enabled. High indicates PLL is locked to input clock. Provides 1× clock output when PLL clock multiplier is disabled. Maximum fanout is one (i.e., <10 pF).
26	RESET	Resets internal divider by bringing momentarily high when PLL is disabled to synchronize internal 1× clock to the input data and/or multiple AD9772A devices.
27, 28	DIV1, DIV0	DIV1 along with DIV0 sets the PLL’s prescaler divide ratio (refer to Table III.)
29	CLK+	Noninverting input to differential clock. Bias to midsupply (i.e., CLKVDD/2).
30	CLK–	Inverting input to differential clock. Bias to midsupply (i.e., CLKVDD/2).
31	CLKCOM	Clock Input Common.
32	CLKVDD	Clock Input Supply Voltage (+2.8V to +3.2V).
33	PLLCOM	Phase Lock Loop Common.
34	PLLVDD	Phase Lock Loop (PLL) Supply Voltage (+3.1V to +3.5V). To disable PLL clock multiplier, connect PLLVDD to PLLCOM.
35	LPF	PLL Loop Filter Node.
36	SLEEP	Power-Down Control Input. Active High. Connect to ACOM if not used.
37, 41, 44	ACOM	Analog Common.
38	REFLO	Reference Ground when Internal 1.2V Reference Used. Connect to AVDD to disable internal reference.
39	REFIO	Reference Input/Output. Serves as reference input when internal reference disabled (i.e., tie REFLO to AVDD). Serves as 1.2V reference output when internal reference activated (i.e., tie REFLO to ACOM). Requires 0.1µF capacitor to ACOM when internal reference activated.
40	FSADJ	Full-Scale Current Output Adjust.
42	I <sub>OUTB</sub>	Complementary DAC Current Output. Full-scale current when all data bits are 0s.
43	I <sub>OUTA</sub>	DAC Current Output. Full-scale current when all data bits are 1s.
45, 46	AVDD	Analog Supply Voltage (+2.8 V to +3.2 V).

## PIN CONFIGURATION



## DEFINITIONS OF SPECIFICATIONS

**Linearity Error (Also Called Integral Nonlinearity or INL)**  
Linearity error is defined as the maximum deviation of the actual analog output from the ideal output, determined by a straight line drawn from zero to full scale.

**Differential Nonlinearity (or DNL)**

DNL is the measure of the variation in analog value, normalized to full scale, associated with a 1 LSB change in digital input code.

**Monotonicity**

A D/A converter is monotonic if the output either increases or remains constant as the digital input increases.

**Offset Error**

The deviation of the output current from the ideal of zero is called offset error. For  $I_{OUTA}$ , 0 mA output is expected when the inputs are all 0s. For  $I_{OUTB}$ , 0 mA output is expected when all inputs are set to 1s.

**Gain Error**

The difference between the actual and ideal output span. The actual span is determined by the output when all inputs are set to 1s, minus the output when all inputs are set to 0s.

**Output Compliance Range**

The range of allowable voltage at the output of a current-output DAC. Operation beyond the maximum compliance limits may cause either output stage saturation or breakdown, resulting in nonlinear performance.

**Temperature Drift**

Temperature drift is specified as the maximum change from the ambient (+25°C) value to the value at either  $T_{MIN}$  or  $T_{MAX}$ . For offset and gain drift, the drift is reported in ppm of full-scale range (FSR) per degree C. For reference drift, the drift is reported in ppm per degree C.

**Power Supply Rejection**

The maximum change in the full-scale output as the supplies are varied from minimum to maximum specified voltages.

**Settling Time**

The time required for the output to reach and remain within a specified error band about its final value, measured from the start of the output transition.

**Glitch Impulse**

Asymmetrical switching times in a DAC give rise to undesired output transients that are quantified by a glitch impulse. It is specified as the net area of the glitch in pV-s.

**Spurious-Free Dynamic Range**

The difference, in dB, between the rms amplitude of the output signal and the peak spurious signal over the specified bandwidth.

**Total Harmonic Distortion**

THD is the ratio of the rms sum of the first six harmonic components to the rms value of the measured fundamental. It is expressed as a percentage or in decibels (dB).

**Signal-to-Noise Ratio (SNR)**

S/N is the ratio of the rms value of the measured output signal to the rms sum of all other spectral components below the Nyquist frequency, excluding the first six harmonics and dc. The value for SNR is expressed in decibels.

**Passband**

Frequency band in which any input applied therein passes unattenuated to the DAC output.

**Stopband Rejection**

The amount of attenuation of a frequency outside the passband applied to the DAC, relative to a full-scale signal applied at the DAC input within the passband.

**Group Delay**

Number of input clocks between an impulse applied at the device input and peak DAC output current.

**Impulse Response**

Response of the device to an impulse applied to the input.

**Adjacent Channel Power Ratio (or ACPR)**

A ratio in dBc between the measured power within a channel relative to its adjacent channel.

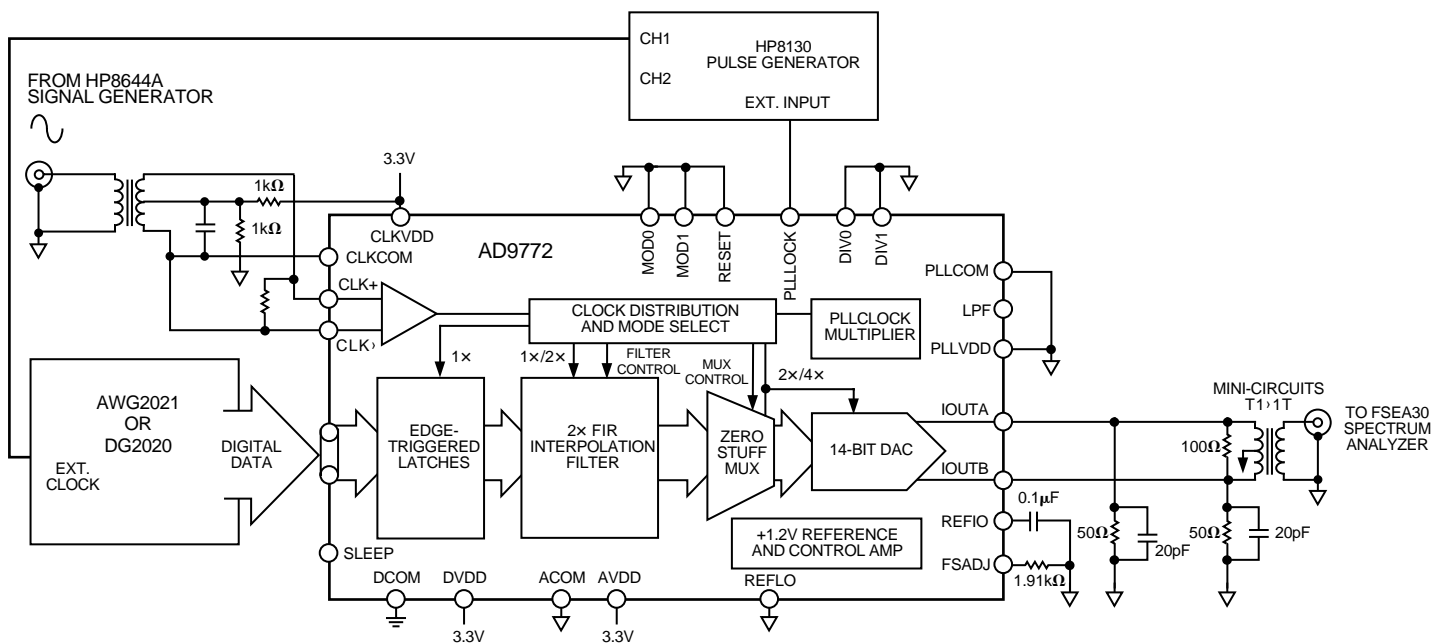


Figure 3. Basic AC Characterization Test Setup



## Typical AC Characterization Curves

(AVDD = +3.3 V, CLKVDD = +3.3 V, PLLVDD = 0 V, DVDD = +3.3 V, I<sub>OUTFS</sub> = 20 mA, PLL Disabled)

Figure 4. Single-Tone Spectral Plot  
@ f<sub>DATA</sub> = 25 MSPS with f<sub>OUT</sub> = f<sub>DATA</sub>/3

Figure 5. "In-Band" SFDR vs. f<sub>OUT</sub>  
@ f<sub>DATA</sub> = 25 MSPS

Figure 6. "Out-of-Band" SFDR vs.  
f<sub>OUT</sub> @ f<sub>DATA</sub> = 25 MSPS

Figure 7. Single-Tone Spectral Plot  
@ f<sub>DATA</sub> = 65 MSPS with f<sub>OUT</sub> = f<sub>DATA</sub>/3

Figure 8. "In-Band" SFDR vs. f<sub>OUT</sub>  
@ f<sub>DATA</sub> = 65 MSPS

Figure 9. "Out-of-Band" SFDR vs.  
f<sub>OUT</sub> @ f<sub>DATA</sub> = 65 MSPS

Figure 10. Single-Tone Spectral Plot  
@ f<sub>DATA</sub> = 160 MSPS with f<sub>OUT</sub> = f<sub>DATA</sub>/3

Figure 11. "In-Band" SFDR vs. f<sub>OUT</sub>  
@ f<sub>DATA</sub> = 160 MSPS

Figure 12. "Out-of-Band" SFDR vs.  
f<sub>OUT</sub> @ f<sub>DATA</sub> = 160 MSPS

PRELIMINARY  
TECHNICAL  
DATA

# AD9772A

Figure 13. "In-Band" Single-Tone SFDR vs.  $A_{OUT}$  @  $f_{OUT} = f_{DATA}/11$

Figure 14 "In-Band" Dual-Tone SFDR vs.  $A_{OUT}$  @  $f_{OUT} = f_{DATA}/11$

Figure 15. "In-Band" Single-Tone SNR vs.  $A_{OUT}$  @  $f_{OUT} = f_{DATA}/11$

Figure 16. "In-Band" Single-Tone SFDR vs.  $A_{OUT}$  @  $f_{OUT} = f_{DATA}/3$

Figure 17. "In-Band" Dual-Tone SFDR vs.  $A_{OUT}$  @  $f_{OUT} = f_{DATA}/3$

Figure 18. "In-Band" Single-Tone SNR vs.  $A_{OUT}$  @  $f_{OUT} = f_{DATA}/3$

Figure 19. "In-Band" Single-Tone SFDR vs. Temperature @  $f_{OUT} = 5$  MHz,  $A_{OUT} = 0$  dBFS

Figure 20. "In-Band" Dual-Tone SFDR vs.  $AVDD$  @  $f_{OUT} = f_{DATA}/4$ ,  $A_{OUT} = 0$  dBFS

Figure 21. "In-Band" Single-Tone SFDR vs. Temperature @  $f_{OUT} = f_{DATA}/11$ ,  $A_{OUT} = 0$  dBFS

PRELIMINARY  
TECHNICAL  
DATA

## FUNCTIONAL DESCRIPTION

Figure 22 shows a simplified block diagram of the AD9772A. The AD9772A is a complete,  $2\times$  oversampling, 14-bit DAC that includes a  $2\times$  interpolation filter, a phase-locked loop (PLL) clock multiplier and a 1.20 V bandgap voltage reference. While the AD9772A's digital interface can support input data rates as high as 160 MSPS, its internal DAC can operate up to 400 MSPS, thus providing direct IF conversion capabilities. The 14-bit DAC provides two complementary current outputs whose full-scale current is determined by an external resistor. The AD9772A features a flexible, low jitter, differential clock input providing excellent noise rejection while accepting a sine wave input. An on-chip PLL clock multiplier produces all of the necessary synchronized clocks from an external reference clock source. Separate supply inputs are provided for each functional block to ensure optimum noise and distortion performance. A SLEEP mode is also included for power savings.

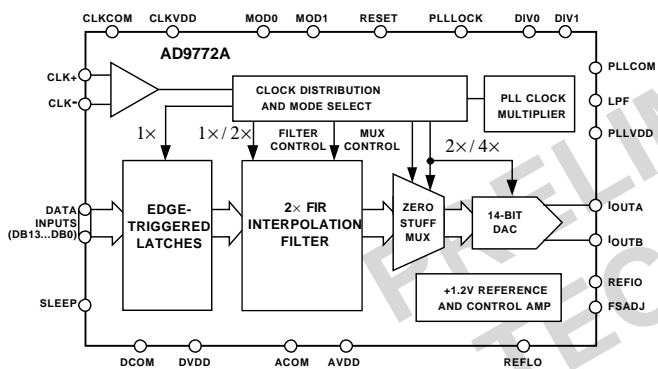


Figure 22. Functional Block Diagram

Preceding the 14-bit DAC is a  $2\times$  digital interpolation filter that can be configured for a low pass (i.e., baseband mode) or high pass (i.e., direct IF mode) response. The input data is latched into the edge-triggered input latches on the rising edge of the differential input clock as shown in Figure 1a and then interpolated by a *factor of two* by the digital filter. For traditional base-band applications, the  $2\times$  interpolation filter has a low pass response. For direct IF applications, the filter's response can be converted into a high pass response to extract the higher image. The output data of the  $2\times$  interpolation filter can update the 14-bit DAC directly or undergo a “zero-stuffing” process to increase the DAC update rate by another *factor of two*. This action enhances the relative signal level and passband flatness of the higher images.

## DIGITAL MODES OF OPERATION

The AD9772A features four different digital modes of operation controlled by the digital inputs, MOD0 and MOD1. MOD0 controls the  $2\times$  digital filter's response (i.e., low pass or high pass), while MOD1 controls the “zero-stuffing” option. The selected mode as shown in Table II will depend on whether the application requires the reconstruction of a baseband or IF signal.

Applications requiring the highest dynamic range over a wide bandwidth should consider operating the AD9772A in a baseband mode. Note, the “zero-stuffing” option can

Table II. Digital Modes

Digital Mode	MOD0	MOD1	Digital Filter	Zero-Stuffing
Baseband	0	0	Low	No
Baseband	0	1	Low	Yes
Direct IF	1	0	High	No
Direct IF	1	1	High	Yes

also be used in this mode although the ratio of signal to image power will be reduced. Applications requiring the synthesis of IF signals should consider operating the AD9772A in a Direct IF mode. In this case, the “zero-stuffing” option should be considered when synthesizing and selecting IFs beyond the input data rate,  $f_{\text{DATA}}$ . If the reconstructed IF falls below  $f_{\text{DATA}}$ , the “zero-stuffing” option may or may not be beneficial. Note, the dynamic range (i.e., SNR/SFDR) is also optimized by disabling the PLL Clock Multiplier (i.e., PLLVDD to PLLCOM) and using an external low jitter clock source operating at the DAC update rate,  $f_{\text{DAC}}$ .

## 2x Interpolation Filter Description

The  $2\times$  interpolation filter is based on a 43-tap half-band symmetric FIR topology that can be configured for a low or high pass response, depending on state of the MOD0 control input. The low pass response is selected with MOD0 LOW while the high pass response is selected with MOD0 HIGH. The low pass frequency and impulse response of the half-band interpolation filter are shown in Figures 2a and 2b, while Table I lists the idealized filter coefficients. Note, a FIR filter's impulse response is also represented by its idealized filter coefficients.

The  $2\times$  interpolation filter essentially multiplies the input data rate to the DAC by a factor of two, relative to its original input data rate, while simultaneously reducing the magnitude of the 1st image associated with the original input data rate occurring at  $f_{\text{DATA}} - f_{\text{FUNDAMENTAL}}$ . Note, as a result of the  $2\times$  interpolation, the digital filter's frequency response is uniquely defined over its Nyquist zone of dc to  $f_{\text{DATA}}$ , with mirror images occurring in adjacent Nyquist zones.

The benefits of an interpolation filter are clearly seen in Figure 23, which shows an example of the frequency and time domain representation of a discrete time sine wave signal before and after it is applied to the  $2\times$  digital interpolation filter in a low pass configuration. Images of the sine wave signal appear around multiples of the DAC's input data rate (i.e.,  $f_{\text{DATA}}$ ) as predicted by sampling theory. These undesirable images will also appear at the output of a reconstruction DAC, although attenuated by the DAC's  $\sin(x)/x$  roll-off response.

In many bandlimited applications, the images from the reconstruction process must be suppressed by an analog filter following the DAC. The complexity of this analog filter is typically determined by the proximity of the desired fundamental to the first image and the required amount of image suppression. Adding to the complexity of this analog filter may be the requirement of compensating for the DAC's  $\sin(x)/x$  response.

Referring to Figure 23, the “new” 1st image associated with the DAC’s higher data rate after interpolation is “pushed” out further relative to the input signal, since it now occurs at  $2 \times f_{\text{DATA}} - f_{\text{FUNDAMENTAL}}$ . The “old” first image associated with the lower DAC data rate before interpolation is suppressed by the digital filter. As a result, the transition band for the analog reconstruction filter is increased, thus reducing the complexity of the analog filter. Furthermore, the  $\sin(x)/x$  roll-off over the original input data passband (i.e., dc to  $f_{\text{DATA}}/2$ ) is significantly reduced.

As previously mentioned, the  $2 \times$  interpolation filter can be converted into a high pass response, thus suppressing the “fundamental” while passing the “original” 1st image

occurring at  $f_{\text{DATA}} - f_{\text{FUNDAMENTAL}}$ . Figure 24 shows the time and frequency representation for a high pass response of a discrete time sine wave. This action can also be modeled as a “1/2 wave” digital mixing process in which the impulse response of the low-pass filter is digitally mixed with a square wave having a frequency of exactly  $f_{\text{DATA}}/2$ . Since the even coefficients have a zero value (refer to Table I), this process simplifies into inverting the center coefficient of the low-pass filter (i.e., invert  $H(18)$ ). Note, this also corresponds into inverting the peak of the impulse response shown in Figure 2a. The resulting high pass frequency response becomes the frequency inverted mirror image of the low-pass filter response shown in Figure 2b.

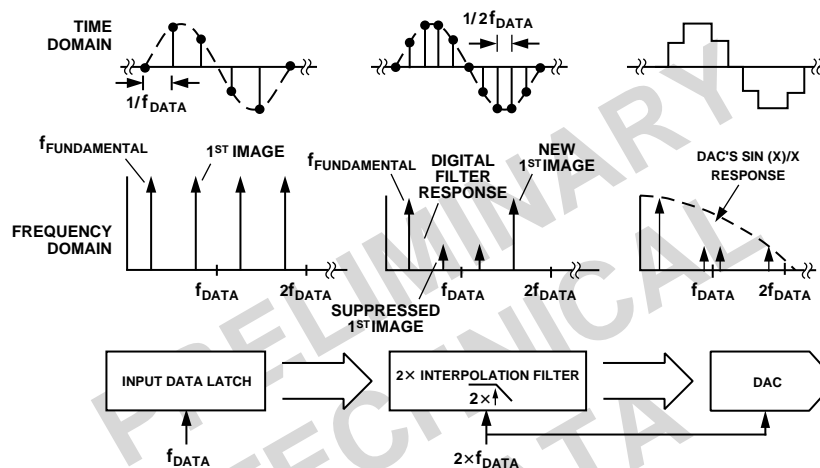


Figure 23. Time and Frequency Domain Example of Low-Pass 2x Digital Interpolation Filter

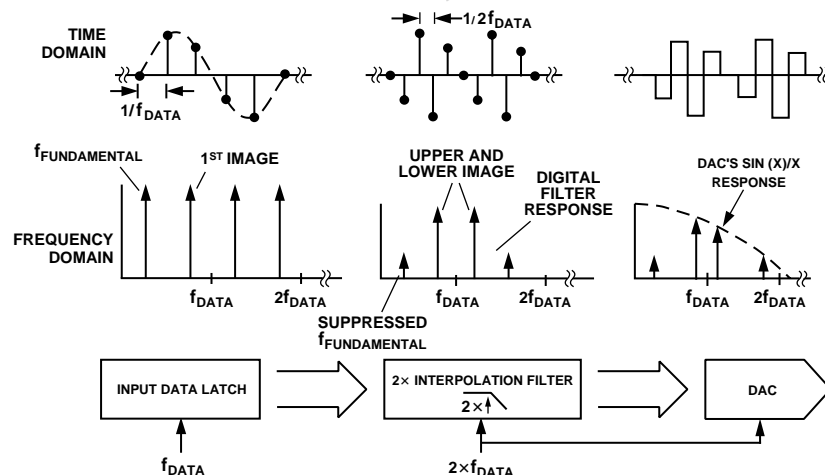


Figure 24. Time and Frequency Domain Example of High-Pass 2x Digital Interpolation Filter

It is worth noting that the “new” 1st image now occurs at  $f_{\text{DATA}} + f_{\text{FUNDAMENTAL}}$ . A reduced transition region of  $2 \times f_{\text{FUNDAMENTAL}}$  exists for image selection, thus mandating that the  $f_{\text{FUNDAMENTAL}}$  be placed sufficiently high for practical filtering purposes in direct IF applications. Also, the “lower sideband images” occurring at  $f_{\text{DATA}} - f_{\text{FUNDAMENTAL}}$  and its multiples (i.e.,  $N \cdot f_{\text{DATA}} - f_{\text{FUNDAMENTAL}}$ ) experience a frequency inversion while the “upper sideband images” occurring at  $f_{\text{DATA}} + f_{\text{FUNDAMENTAL}}$  and its multiples (i.e.,  $N \cdot f_{\text{DATA}} + f_{\text{FUNDAMENTAL}}$ ) do not.

### “Zero Stuffing” Option Description

As shown in Figure 25, a “zero” or null in the frequency responses (after interpolation and DAC reconstruction) occurs at the final DAC update rate (i.e.,  $2 \times f_{\text{DATA}}$ ) due to the DAC’s inherent  $\sin(x)/x$  roll-off response. In baseband applications, this roll-off in the frequency response may not be as problematic since much of the desired signal energy remains below  $f_{\text{DATA}}/2$  and the amplitude variation is not as severe. However, in direct IF applications interested in extracting an image above  $f_{\text{DATA}}/2$ , this roll-off may be problematic due to the increased passband amplitude variation as well as the reduced signal level of the higher images.

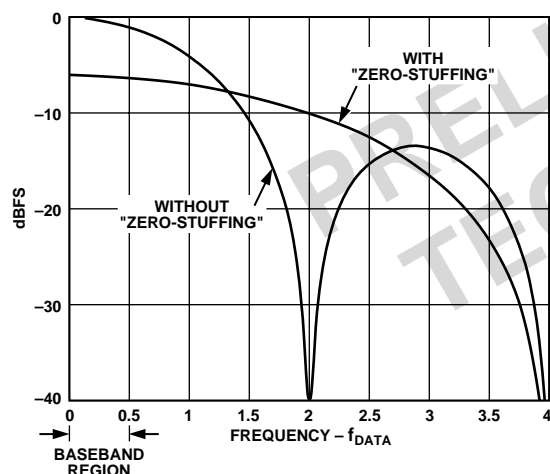


Figure 25. Effects “Zero-Stuffing” on DAC’s  $\sin(x)/x$  Response

For instance, if the digital data into the AD9772A represented a baseband signal centered around  $f_{\text{DATA}}/4$  with a passband of  $f_{\text{DATA}}/10$ , the reconstructed baseband signal out of the AD9772A would experience only a 0.18 dB amplitude variation over its passband with the “1st image” occurring at  $7/4 f_{\text{DATA}}$  with 17 dB of attenuation relative to the fundamental. However, if the high-pass filter response was selected, the AD9772A would now produce pairs of images at  $[(2N + 1) \times f_{\text{DATA}}] \pm f_{\text{DATA}}/4$  where  $N = 0, 1 \dots$ . Note, due to the DAC’s  $\sin(x)/x$  response, only the lower or upper sideband images centered around  $f_{\text{DATA}}$  may be useful although they would be attenuated by  $-2.1$  dB and  $-6.54$  dB respectively as well as experience a passband amplitude roll-off of 0.6 dB and 1.3 dB.

To improve upon the passband flatness of the desired image and/or to extract higher images (i.e.,  $3 \times f_{\text{DATA}} \pm f_{\text{FUNDAMENTAL}}$ ) the “zero-stuffing” option should be employed by bringing the MOD1 pin HIGH. This option

increases the effective DAC update rate by another *factor of two* since a “midscale” sample (i.e., 10 0000 0000 0000) is inserted after every data sample originating from the  $2 \times$  interpolation filter. A digital multiplexer switching at a rate of  $4 \times f_{\text{DATA}}$  between the interpolation filter’s output and a data register containing the “midscale” data sample is used to implement this option as shown in Figure 24. Hence, the DAC output is now forced to return to its differential midscale current value (i.e.,  $I_{\text{OUTA}} - I_{\text{OUTB}} \equiv 0$  mA) after reconstructing each data sample from the digital filter.

The net effect is to increase the DAC update rate such that the “zero” in the  $\sin(x)/x$  frequency response now occurs at  $4 \times f_{\text{DATA}}$  along with a corresponding reduction in output power as shown in Figure 25. Note, if the  $2 \times$  interpolation filter’s high pass response is also selected, this action can be modeled as a “1/4 wave” digital mixing process since this is equivalent to digitally mixing the impulse response of the low-pass filter with a square wave having a frequency of exactly  $f_{\text{DATA}}$  (i.e.,  $f_{\text{DAC}}/4$ ).

It is important to realize that the “zero stuffing” option by itself does not change the location of the images but rather their signal level, amplitude flatness and relative weighting. For instance, in the previous example, the passband amplitude flatness of the lower and upper sideband images centered around  $f_{\text{DATA}}$  are improved to 0.14 dB and 0.24 dB respectively, while the signal level has changed to  $-6.5$  dBFS and  $-7.5$  dBFS. The lower or upper sideband image centered around  $3 \times f_{\text{DATA}}$  will exhibit an amplitude flatness of 0.77 dB and 1.29 dB with signal levels of approximately  $-14.3$  dBFS and  $-19.2$  dBFS.

### PLL CLOCK MULTIPLIER OPERATION

The Phase Lock Loop (PLL) clock multiplier circuitry along with the clock distribution circuitry can produce the necessary internally synchronized  $1 \times$ ,  $2 \times$ , and  $4 \times$  clocks for the edge triggered latches,  $2 \times$  interpolation filter, “zero stuffing” multiplier, and DAC. Figure 26 shows a functional block diagram of the PLL clock multiplier, which consists of a phase detector, a charge pump, a voltage controlled oscillator (VCO), a prescaler, and digital control inputs/outputs. The clock distribution circuitry generates all the internal clocks for a given mode of operation. The charge pump and VCO are powered from PLLVDD while the differential clock input buffer, phase detector, prescaler and clock distribution circuitry are powered from CLKVDD. To ensure optimum phase noise performance from the PLL clock multiplier and clock distribution circuitry, PLLVDD and CLKVDD must originate from the same clean analog supply.

The PLL clock multiplier has two modes of operation. It can be enabled for less demanding applications providing a reference clock meeting the minimum specified input data rate of 6 MSPS. It can be disabled for applications below this data rate or for applications requiring higher phase noise performance. In this case, a reference clock at *twice* the input data rate (i.e.,  $2 \times f_{\text{DATA}}$ ) must be provided *without* the “zero stuffing” option selected and *four times* the input data rate (i.e.,  $4 \times f_{\text{DATA}}$ ) *with* the “zero stuffing” option selected. Note, multiple AD9772A devices can be synchro-

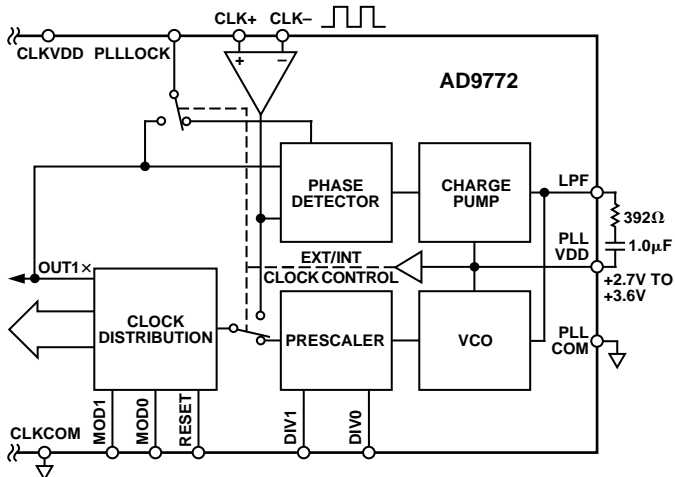


Figure 26. Clock Multiplier with PLL Clock Multiplier Enabled

nized in either mode if driven by the same reference clock since the PLL clock multiplier when enabled ensures synchronization. RESET can be used for synchronization if the PLL clock multiplier is disabled.

Figure 26 shows the proper configuration used to enable the PLL clock multiplier. In this case, the external clock source is applied to CLK+ (and/or CLK-) and the PLL clock multiplier is fully enabled by connecting PLLVDD to CLKVDD. An external PLL loop filter consisting of a series resistor and ceramic capacitor connected from the output of the charge pump (i.e., LPF) to PLLVDD is required for stability of the PLL. Also, a shield surrounding these components is recommended to minimize external noise coupling into the VCO input.

The components values shown (i.e., 392  $\Omega$  and 1.0  $\mu\text{F}$ ) were selected to optimize the phase noise vs. settling/acquisition time characteristics of the PLL. The settling/acquisition time characteristics are also dependent on the divide-by-N ratio as well as the input data rate. In general, the acquisition time increases with increasing data rate (for fixed divide-by-N ratio) or increasing divide-by-N ratio (for fixed input data rate).

Since the VCO can operate over a 96 MHz–400 MHz range, the prescaler divide-by-ratio following the VCO must be set according to Table III for a given input data rate (i.e.,  $f_{\text{DATA}}$ ) to ensure optimum phase noise and successful “locking.” In general, the best phase noise performance for any prescaler setting is achieved with the VCO operating near its maximum output frequency of 400 MHz. Note, the divide-by-N ratio also depends on whether the “zero stuffing” option is enabled since this option requires the DAC to operate at four times the input data rate. The divide-by-N ratio is set by DIV1 and DIV0.

With the PLL clock multiplier enabled, PLLLOCK serves as an active HIGH control output which may be monitored upon system power-up to indicate that the PLL is successfully “locked” to the input clock. Note, when the PLL clock multiplier is NOT locked, PLLLOCK will toggle between logic HIGH and LOW in an asynchronous man-

Table III. Recommended Prescaler Divide-by-N Ratio Settings

$f_{\text{DATA}}$ (MSPS)	MOD1	DIV1	DIV0	Divide-by-N Ratio
48-160	0	0	0	1
24-100	0	0	1	2
12-50	0	1	0	4
6-25	0	1	1	8
24-100	1	0	0	1
12-50	1	0	1	2
6-25	1	1	0	4
3-12.5	1	1	1	8

ner until locking is finally achieved. As a result, it is recommended that PLLLOCK, if monitored, be sampled several times to detect proper locking 100 ms upon power-up.

As stated earlier, applications requiring input data rates below 6 MSPS must disable the PLL clock multiplier and provide an external reference clock. However, applications already containing a low phase noise (i.e., jitter) reference clock that is *twice (or four times)* the input data rate should consider disabling the PLL clock multiplier to achieve the best SNR performance from the AD9772. Note, the SFDR performance and wideband noise performance of the AD9772A remains unaffected with or without the PLL clock multiplier enabled.

The effects of phase noise on the AD9772's SNR performance becomes more noticeable at higher reconstructed output frequencies and signal levels. Figure 27 compares the phase noise of a full-scale sine wave at exactly  $f_{\text{DATA}}/4$  at different data rates (hence carrier frequency) with the optimum DIV1, DIV0 setting. The effects of phase noise, and its effect on a signal's CNR performance, becomes even more evident at higher IF frequencies as shown in Figure 28. In both instances, it is the “narrowband” phase noise that limits the CNR performance.

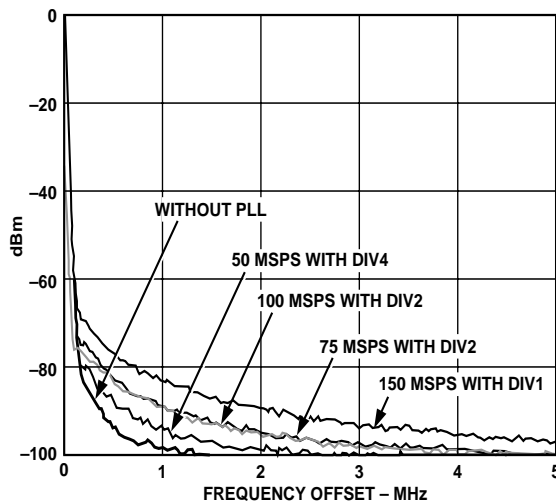


Figure 27. Phase Noise of PLL Clock Multiplier @ Exactly  $f_{OUT} = f_{DATA}/4$  at Different  $f_{DATA}$  Settings with Optimum DIV0/DIV1 Settings Using R & S FSEA30 SpectrumAnalyzer

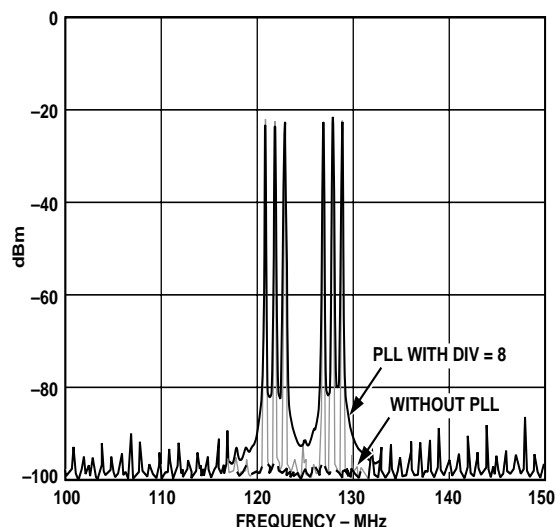


Figure 28. Direct IF Mode Reveals Phase Noise Degradation With and Without PLL Clock Multiplier (IF = 125 MHz and  $f_{DATA} = 100$  MSPS)

To disable the PLL Clock Multiplier, connect PLLVDD to PLLCOM as shown in Figure 29. LPF may remain open since this portion of the PLL circuitry is now disabled. The differential clock input should be driven with a reference clock *twice* the data input rate in baseband applications and *four* time the data input rate in direct IF applications in which the “1/4 wave” mixing option is employed (i.e., MOD1 and MOD0 active HIGH). The clock distribution circuitry remains enabled providing a 1× internal clock at PLLLOCK. Digital input data is latched into the AD9772 on every other rising edge of the differential clock input. The rising edge which corresponds to the input latch immediately precedes the rising edge of the 1× clock at PLLLOCK. Adequate setup and hold time for the input data as shown in Figure 1b should be allowed. Note that enough delay is present between CLK+/CLK- and the data input latch to cause the minimum set-up time for input data to be negative. This is noted in DIGITAL SPECIFICATIONS on page 4. PLLLOCK contains a

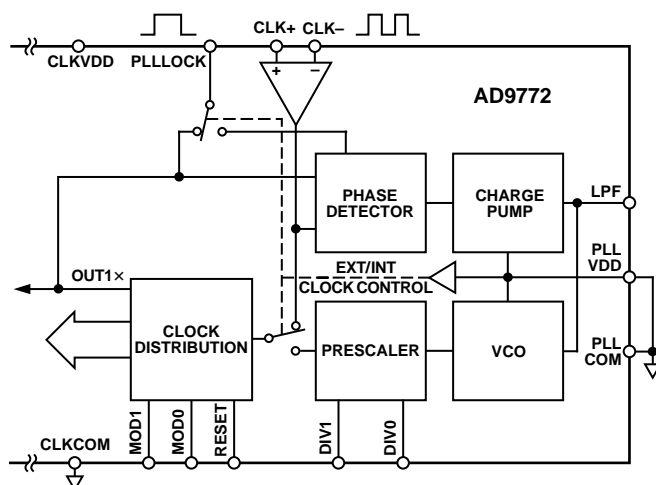


Figure 29. Clock Multiplier with PLL CLOCK Multiplier Disabled

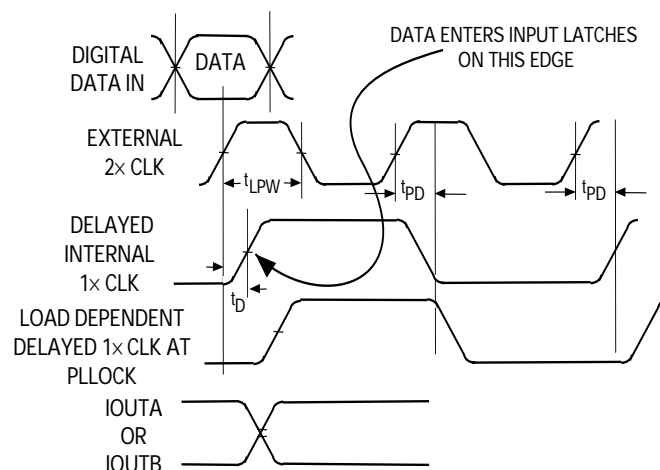


Figure 30. Internal Timing of AD9772A with PLL Disabled

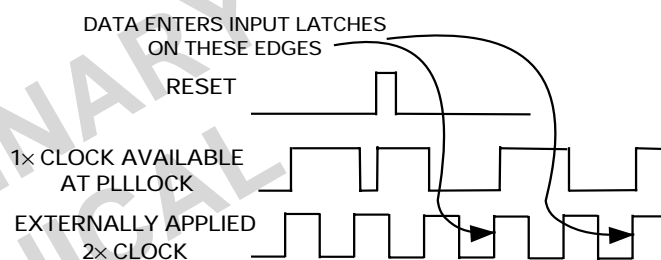


Figure 31. RESET Timing of AD9772A with PLL Disabled

relatively weak driver output, with its output delay ( $t_{OD}$ ) sensitive to output capacitance loading. Thus PLLLOCK should be buffered for fanouts greater than one and/or load capacitance greater than 10 pF. If a data timing issue exists between the AD9772A and its external driver device, the 1× clock appearing at PLLLOCK can be inverted via an external gate to ensure proper setup and hold time.

### SYNCHRONIZATION OF CLK/DATA USING RESET WITH PLL DISABLED

The relationship between the internal and external clocks in this mode is shown in Figure 30. A clock at the output update data rate (2× the input data rate) must be applied to the CLK inputs. Internal dividers create the internal 1× clock necessary for the input latches. With the PLL disabled, a delayed version of the 1× clock is present at the PLLLOCK pin. The DAC latch is updated on the particular rising edge of the external 2× clock which corresponds to the rising edge of the 1× clock. Updates to the input data should be synchronized to this specific rising edge as shown in Figure 30. To ensure this synchronization, a logic “1” should be momentarily applied to the RESET pin on power up, before CLK is applied. Applying a momentary logic “1” to RESET brings the 1× clock at PLLLOCK to a logic “1”. On the next rising edge of the 2× clock, the 1× clock will go to logic “0”. The following rising edge of the 2× clock will cause the 1× clock to logic “1” again, as well as update the data in both of the input latches. The details of this are given in figure 31.

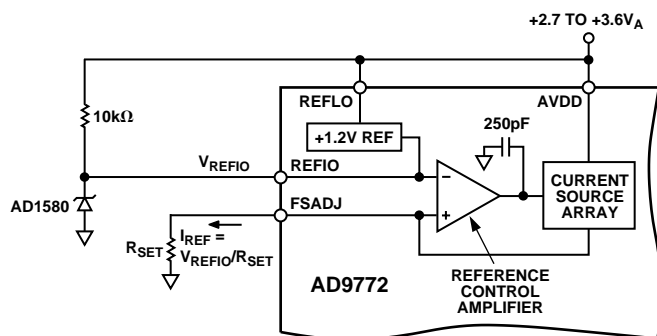


Figure 32. Block Diagram of Internal DAC, 1.2 V Reference, and Reference Control Circuits

## DAC OPERATION

The 14-bit DAC along with the 1.2 V reference and reference control amplifier is shown in Figure 32. The DAC consists of a large PMOS current source array capable of providing up to 20 mA of full-scale current,  $I_{OUTFS}$ . The array is divided into thirty-one equal currents that make up the five most significant bits (MSBs). The next four bits, or middle bits, consist of 15 equal current sources whose values are 1/16th of an MSB current source. The remaining LSBs are binary weighted fractions of the middle-bits' current sources. All of these current sources are switched to one or the other of two output nodes (i.e.,  $I_{OUTA}$  or  $I_{OUTB}$ ) via PMOS differential current switches. Implementing the middle and lower bits with current sources, instead of an R-2R ladder, enhances its dynamic performance for multitone or low amplitude signals and helps maintain the DAC's high output impedance. The full-scale output current is regulated by the reference control amplifier and can be set from 2 mA to 20 mA via an external resistor,  $R_{SET}$ . The external resistor, in combination with both the reference control amplifier and voltage reference,  $REFIO$ , sets the reference current,  $I_{REF}$ , which is mirrored over to the segmented current sources with the proper scaling factor. The full-scale current,  $I_{OUTFS}$ , is exactly thirty-two times the value of  $I_{REF}$ .

## DAC TRANSFER FUNCTION

The AD9772A provides complementary current outputs,  $I_{OUTA}$  and  $I_{OUTB}$ .  $I_{OUTA}$  will provide a near full-scale current output,  $I_{OUTFS}$ , when all bits are high (i.e., DAC CODE = 16383) while  $I_{OUTB}$ , the complementary output, provides no current. The current output appearing at  $I_{OUTA}$  and  $I_{OUTB}$  is a function of both the input code and  $I_{OUTFS}$  and can be expressed as:

$$I_{OUTA} = (DAC\ CODE/16384) \times I_{OUTFS} \quad (1)$$

$$I_{OUTB} = (16383 - DAC\ CODE)/16384 \times I_{OUTES} \quad (2)$$

where *DAC CODE* = 0 to 16383 (i.e., Decimal Representation).

As previously mentioned,  $I_{OUTFS}$  is a function of the reference current  $I_{REF}$ , which is nominally set by a reference voltage  $V_{REFIO}$ , and external resistor,  $R_{SET}$ . It can be expressed as:

$$I_{OUTES} = 32 \times I_{REF} \quad (3)$$

where

$$I_{REF} = V_{REFIO}/R_{SET} \quad (4)$$

The two current outputs will typically drive a resistive load directly or via a transformer. If dc coupling is required,  $I_{OUTA}$  and  $I_{OUTB}$  should be directly connected to matching resistive loads,  $R_{LOAD}$ , that are tied to analog common, ACOM. Note that  $R_{LOAD}$  may represent the equivalent load resistance seen by  $I_{OUTA}$  or  $I_{OUTB}$  as would be the case in a doubly terminated 50  $\Omega$  or 75  $\Omega$  cable. The single-ended voltage output appearing at the  $I_{OUTA}$  and  $I_{OUTB}$  nodes is simply:

$$V_{OUTA} = I_{OUTA} \times R_{LOAD} \quad (5)$$

$$V_{OUTB} = I_{OUTB} \times R_{LOAD} \quad (6)$$

Note that the full-scale value of  $V_{OUTA}$  and  $V_{OUTB}$  should not exceed the specified output compliance range of 1.25 V to prevent signal compression. To maintain optimum distortion and linearity performance, the maximum voltages at  $V_{OUTA}$  and  $V_{OUTB}$  should not exceed  $\pm 500$  mV p-p.

The differential voltage,  $V_{\text{DIFF}}$ , appearing across  $I_{\text{OUTA}}$  and  $I_{\text{OUTB}}$ , is:

$$V_{DIFF} = (I_{OUTA} - I_{OUTB}) \times R_{LOAD} \quad (7)$$

Substituting the values of  $I_{OUTA}$ ,  $I_{OUTB}$  and  $I_{REF}$ ;  $V_{DIFF}$  can be expressed as:

$$V_{DIFF} = [(2 \text{ DAC CODE} - 16383)/16384] \times (32 R_{LOAD}/R_{SET}) \times V_{REFIO} \quad (8)$$

The last two equations highlight some of the advantages of operating the AD9772A differentially. First, the differential operation will help cancel common-mode error sources such as noise, distortion and dc offsets associated with  $I_{OUTA}$  and  $I_{OUTB}$ . Second, the differential code-dependent current and subsequent voltage,  $V_{DIFF}$ , is twice the value of the single-ended voltage output (i.e.,  $V_{OUTA}$  or  $V_{OUTB}$ ), thus providing twice the signal power to the load.

Note that the gain drift temperature performance for a single-ended ( $V_{OUTA}$  and  $V_{OUTB}$ ) or differential output ( $V_{DIFF}$ ) of the AD9772A can be enhanced by selecting temperature tracking resistors for  $R_{LOAD}$  and  $R_{SET}$  due to their ratiometric relationship as shown in Equation 8.

## REFERENCE OPERATION

The AD9772A contains an internal 1.20 V bandgap reference that can easily be disabled and overridden by an external reference. REFIO serves as either an *output* or *input*, depending on whether the internal or external reference is selected. If REFLO is tied to ACOM, as shown in Figure 33, the internal reference is activated, and REFIO provides a 1.20 V output. In this case, the internal reference *must* be compensated externally with a ceramic chip capacitor of 0.1  $\mu\text{F}$  or greater from REFIO to REFLO. If any additional loading is required, REFIO should be buffered with an external amplifier having an input bias current less than 100 nA.



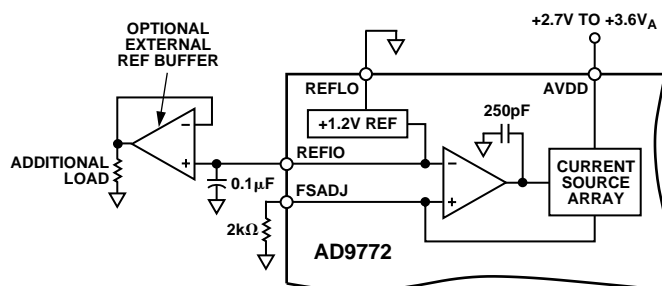


Figure 33. Internal Reference Configuration

The internal reference can be disabled by connecting REFLO to AVDD. In this case, an external 1.2 V reference such as the AD1580 may then be applied to REFIO as shown in Figure 34. The external reference may provide either a fixed reference voltage to enhance accuracy and drift performance or a varying reference voltage for gain control. Note that the 0.1  $\mu\text{F}$  compensation capacitor is not required since the internal reference is disabled, and the high input impedance of REFIO minimizes any loading of the external reference.

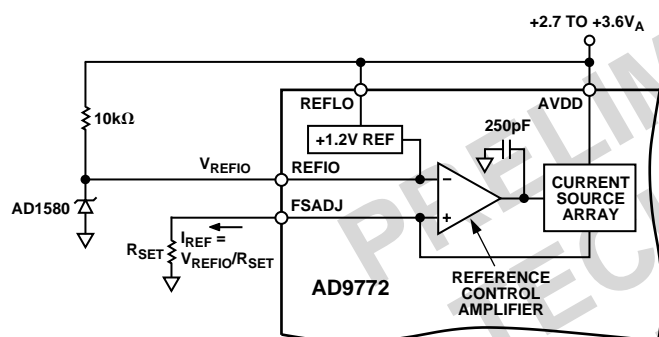


Figure 34. External Reference Configuration

### REFERENCE CONTROL AMPLIFIER

The AD9772A also contains an internal control amplifier that is used to regulate the DAC's full-scale output current,  $I_{\text{OUTFS}}$ . The control amplifier is configured as a V-I converter, as shown in Figure 34, such that its current output,  $I_{\text{REF}}$ , is determined by the ratio of the  $V_{\text{REFIO}}$  and an external resistor,  $R_{\text{SET}}$ , as stated in Equation 4.  $I_{\text{REF}}$  is copied to the segmented current sources with the proper scaling factor to set  $I_{\text{OUTFS}}$  as stated in Equation 3.

The control amplifier allows a wide (10:1) adjustment span of  $I_{\text{OUTFS}}$  over a 2 mA to 20 mA range by setting  $I_{\text{REF}}$  between 62.5  $\mu\text{A}$  and 625  $\mu\text{A}$ . The wide adjustment span of  $I_{\text{OUTFS}}$  provides several application benefits. The first benefit relates directly to the power dissipation of the AD9772's DAC, which is proportional to  $I_{\text{OUTFS}}$  (refer to the Power Dissipation section). The second benefit relates to the 20 dB adjustment, which is useful for system gain control purposes.

$I_{\text{REF}}$  can be controlled using the single-supply circuit shown in Figure 35 for a fixed  $R_{\text{SET}}$ . In this example, the internal reference is disabled, and the voltage of REFIO is varied over its compliance range of 1.25 V to 0.10 V. REFIO can be driven by a single-supply DAC or digital potentiometer, thus allowing  $I_{\text{REF}}$  to be digitally controlled

for a fixed  $R_{\text{SET}}$ . This particular example shows the AD5220, an 8-bit serial input digital potentiometer, along with the AD1580 voltage reference. Note, since the input impedance of REFIO does interact and load the digital potentiometer wiper to create a slight nonlinearity in the programmable voltage divider ratio, a digital potentiometer with 10 k $\Omega$  or less of resistance is recommended.

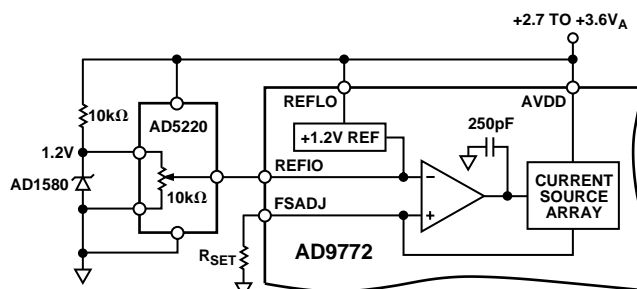


Figure 35. Single-Supply Gain Control Circuit

### ANALOG OUTPUTS

The AD9772A produces two complementary current outputs,  $I_{\text{OUTA}}$  and  $I_{\text{OUTB}}$ , which may be configured for single-ended or differential operation.  $I_{\text{OUTA}}$  and  $I_{\text{OUTB}}$  can be converted into complementary single-ended voltage outputs,  $V_{\text{OUTA}}$  and  $V_{\text{OUTB}}$ , via a load resistor,  $R_{\text{LOAD}}$ , as described in the DAC Transfer Function section, by Equations 5 through 8. The differential voltage,  $V_{\text{DIFF}}$ , existing between  $V_{\text{OUTA}}$  and  $V_{\text{OUTB}}$ , can also be converted to a single-ended voltage via a transformer or differential amplifier configuration.

Figure 36 shows the equivalent analog output circuit of the AD9772A, consisting of a parallel combination of PMOS differential current switches associated with each segmented current source. The output impedance of  $I_{\text{OUTA}}$  and  $I_{\text{OUTB}}$  is determined by the equivalent parallel combination of the PMOS switches and is typically 200 k $\Omega$  in parallel with 3 pF. Due to the nature of a PMOS device, the output impedance is also slightly dependent on the output voltage (i.e.,  $V_{\text{OUTA}}$  and  $V_{\text{OUTB}}$ ) and, to a lesser extent, the analog supply voltage, AVDD, and full-scale current,  $I_{\text{OUTFS}}$ . Although the output impedance's signal dependency can be a source of dc nonlinearity and ac linearity (i.e., distortion), its effects can be limited if certain precautions are noted.

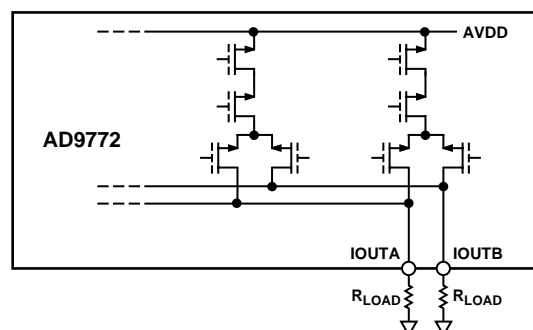


Figure 36. Equivalent Analog Output Circuit

$I_{OUTA}$  and  $I_{OUTB}$  also have a negative and positive voltage compliance range. The negative output compliance range of  $-1.0$  V is set by the breakdown limits of the CMOS process. Operation beyond this maximum limit may result in a breakdown of the output stage and affect the reliability of the AD9772A. The positive output compliance range is slightly dependent on the full-scale output current,  $I_{OUTFS}$ . Operation beyond the positive compliance range will induce clipping of the output signal, which severely degrades the AD9772A's linearity and distortion performance.

Operating the AD9772A with reduced voltage output swings at  $I_{OUTA}$  and  $I_{OUTB}$  in a differential or single-ended output configuration reduces the signal dependency of its output impedance, thus enhancing distortion performance. Although the voltage compliance range of  $I_{OUTA}$  and  $I_{OUTB}$  extends from  $-1.0$  V to  $+1.25$  V, optimum distortion performance is achieved when the maximum full-scale signal at  $I_{OUTA}$  and  $I_{OUTB}$  does not exceed approximately  $0.5$  V. A properly selected transformer with a grounded center-tap will allow the AD9772A to provide the required power and voltage levels to different loads while maintaining reduced voltage swings at  $I_{OUTA}$  and  $I_{OUTB}$ . DC-coupled applications requiring a differential or single-ended output configuration should size  $R_{LOAD}$  accordingly. Refer to Applying the AD9772A section for examples of various output configurations.

The most significant improvement in the AD9772A's distortion and noise performance is realized using a differential output configuration. The common-mode error sources of both  $I_{OUTA}$  and  $I_{OUTB}$  can be substantially reduced by the common-mode rejection of a transformer or differential amplifier. These common-mode error sources include even-order distortion products and noise. The enhancement in distortion performance becomes more significant as the reconstructed waveform's frequency content increases and/or its amplitude decreases. The distortion and noise performance of the AD9772A is also dependent on the full-scale current setting,  $I_{OUTFS}$ . Although  $I_{OUTFS}$  can be set between  $2$  mA and  $20$  mA, selecting an  $I_{OUTFS}$  of  $20$  mA will provide the best distortion and noise performance.

In summary, the AD9772A achieves the optimum distortion and noise performance under the following conditions:

1. Positive voltage swing at  $I_{OUTA}$  and  $I_{OUTB}$  limited to  $+0.5$  V.
2. Differential Operation.
3.  $I_{OUTFS}$  set to  $20$  mA.
4. PLL Clock Multiplier Disabled

Note the majority of the AC Characterization Curves for the AD9772A are performed under the above-mentioned operating conditions.

## DIGITAL INPUTS/OUTPUTS

The AD9772A consists of several digital input pins used for data, clock and control purposes. It also contains a single digital output pin, PLLLOCK, used to monitor the status of the internal PLL clock multiplier or provide a  $1\times$

clock output. The 14-bit parallel data inputs follow standard positive binary coding where DB13 is the most significant bit (MSB), and DB0 is the least significant bit (LSB).  $I_{OUTA}$  produces a full-scale output current when all data bits are at Logic 1.  $I_{OUTB}$  produces a complementary output with the full-scale current split between the two outputs as a function of the input code.

The digital interface is implemented using an edge-triggered master slave latch and is designed to support an input data rate as high as  $160$  MSPS. The clock can be operated at any duty cycle that meets the specified latch pulsewidth as shown in Figures 1a and 1b. The setup and hold times can also be varied within the clock cycle as long as the specified minimum times are met. The digital inputs (excluding CLK+ and CLK-) are CMOS-compatible with its logic thresholds,  $V_{THRESHOLD}$ , set to approximately half the digital positive supply (i.e., DVDD or CLKVDD) or

$$V_{THRESHOLD} = DVDD/2 (\pm 20\%)$$

The internal digital circuitry of the AD9772A is capable of operating over a digital supply range of  $2.8$  V to  $3.2$  V. As a result, the digital inputs can also accommodate TTL levels when DVDD is set to accommodate the maximum high level voltage of the TTL drivers  $V_{OH(MAX)}$ . Although a DVDD of  $3.3$  V will typically ensure proper compatibility with most TTL logic families, a series  $200\ \Omega$  resistors are recommended between the TTL logic driver and digital inputs to limit the peak current through the ESD protection diodes if  $V_{OH(MAX)}$  exceeds DVDD by more than  $300$  mV. Figure 37 shows the equivalent digital input circuit for the data and control inputs.

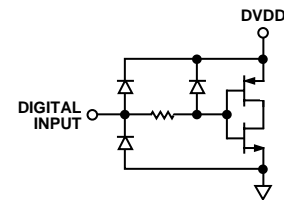


Figure 37. Equivalent Digital Input

The AD9772A features a flexible differential clock input operating from separate supplies (i.e., CLKVDD, CLKCOM) to achieve optimum jitter performance. The two clock inputs, CLK+ and CLK-, can be driven from a single-ended or differential clock source. For single-ended operation, CLK+ should be driven by a single-ended logic source while CLK- should be set to the logic source's threshold voltage via a resistor divider/capacitor network referenced to CLKVDD as shown in Figure 38. For differential operation, both CLK+ and CLK- should be biased to  $CLKVDD/2$  via a resistor divider network as shown in Figure 39. An RF transformer as shown in Figure 3 can also be used to convert a single-ended clock input to a differential clock input.

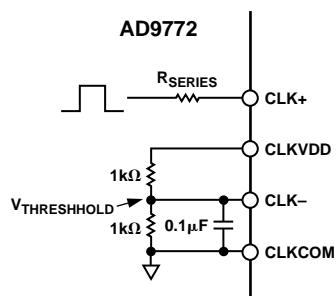


Figure 38. Single-Ended Clock Interface

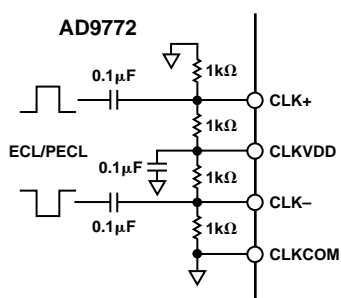


Figure 39. Differential Clock Interface

The quality of the clock and data input signals are important in achieving the optimum performance. The external clock driver circuitry should provide the AD9772A with a low jitter clock input which meets the min/max logic levels while providing fast edges. Although fast clock edges help minimize any jitter that will manifest itself as phase noise on a reconstructed waveform, the high gain-bandwidth product of the AD9772A's differential comparator can tolerate sine wave inputs as low as 0.5 V p-p, with minimal degradation in its output noise floor.

Digital signal paths should be kept short and run lengths matched to avoid propagation delay mismatch. The insertion of a low value resistor network (i.e., 50 Ω to 200 Ω) between the AD9772A digital inputs and driver outputs may be helpful in reducing any overshooting and ringing at the digital inputs that contribute to data feedthrough.

### SLEEP MODE OPERATION

The AD9772A has a SLEEP function that turns off the output current and reduces the analog supply current to less than 6 mA over the specified supply range of 2.8 V to 3.2 V. This mode can be activated by applying a Logic Level "1" to the SLEEP pin. The AD9772A takes less than 50 ns to power down and approximately 15 μs to power back up.

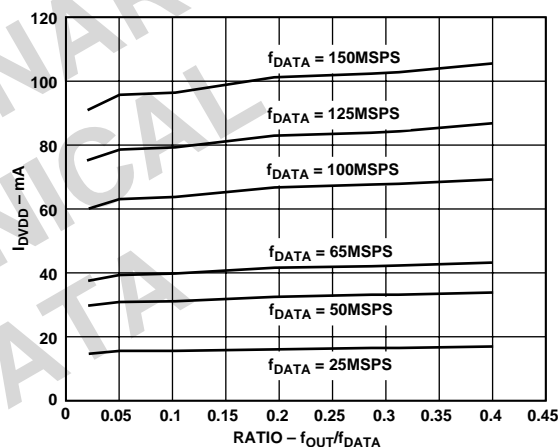
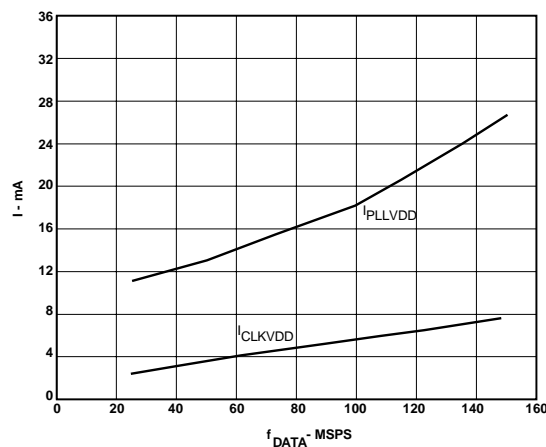
### POWER DISSIPATION

The power dissipation,  $P_D$ , of the AD9772A is dependent on several factors, including:

1. AVDD, PLLVDD, CLKVDD and DVDD, the power supply voltages
2.  $I_{OUTFS}$ , the full-scale current output
3.  $f_{DATA}$ , the update rate
4. the reconstructed digital input waveform.

The power dissipation is directly proportional to the analog supply current,  $I_{AVDD}$ , and the digital supply current,  $I_{DVDD}$ .  $I_{AVDD}$  is directly proportional to  $I_{OUTFS}$ , and is insensitive to  $f_{DATA}$ .

Conversely,  $I_{DVDD}$  is dependent on both the digital input waveform and  $f_{DATA}$ . Figure 40 shows  $I_{DVDD}$  as a function of full-scale sine wave output ratios ( $f_{OUT}/f_{DATA}$ ) for various update rates with DVDD = 3 V. The supply current from CLKVDD and PLLVDD is relatively insensitive to the digital input waveform, but shown directly proportional to the update rate as shown in Figure 41.

Figure 40.  $I_{DVDD}$  vs. Ratio @DVDD=3.3VFigure 41.  $I_{PLLVD}$  and  $I_{CLKVD}$  vs.  $f_{DATA}$

## APPLYING THE AD9772 OUTPUT CONFIGURATIONS

The following sections illustrate some typical output configurations for the AD9772A. Unless otherwise noted, it is assumed that  $I_{OUTFS}$  is set to a nominal 20 mA for optimum performance. For applications requiring the optimum dynamic performance, a differential output configuration is highly recommended. A differential output configuration may consist of either an RF transformer or a differential op amp configuration. The transformer configuration provides the optimum high frequency performance and is recommended for any application allowing for ac coupling. The differential op amp configuration is suitable for applications requiring dc coupling, a bipolar output, signal gain and/or level shifting.

A single-ended output is suitable for applications requiring a unipolar voltage output. A positive unipolar output voltage will result if  $I_{OUTA}$  and/or  $I_{OUTB}$  is connected to an appropriately sized load resistor,  $R_{LOAD}$ , referred to ACOM. This configuration may be more suitable for a single-supply system requiring a dc-coupled, ground-referred output voltage. Alternatively, an amplifier could be configured as an I-V converter, thus converting  $I_{OUTA}$  or  $I_{OUTB}$  into a negative unipolar voltage. This configuration provides the best dc linearity since  $I_{OUTA}$  or  $I_{OUTB}$  is maintained at a virtual ground.

## DIFFERENTIAL COUPLING USING A TRANSFORMER

An RF transformer can be used to perform a differential-to-single-ended signal conversion as shown in Figure 42. A differentially-coupled transformer output provides the optimum distortion performance for output signals whose spectral content lies within the transformer's pass-band. An RF transformer such as the Mini-Circuits T1-1T provides excellent rejection of common-mode distortion (i.e., even-order harmonics) and noise over a wide frequency range. It also provides electrical isolation and the ability to deliver twice the power to the load. Transformers with different impedance ratios may also be used for impedance matching purposes. Note that the transformer provides ac coupling only and its linearity performance degrades at the low end of its frequency range due to core saturation.

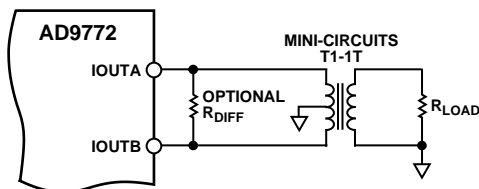


Figure 42. Differential Output Using a Transformer

The center tap on the primary side of the transformer must be connected to ACOM to provide the necessary dc current path for both  $I_{OUTA}$  and  $I_{OUTB}$ . The complementary voltages appearing at  $I_{OUTA}$  and  $I_{OUTB}$  (i.e.,  $V_{OUTA}$  and  $V_{OUTB}$ ) swing symmetrically around ACOM and should be maintained with the specified output compliance range of the AD9772A. A differential resistor,  $R_{DIFF}$ , may be inserted in applications in which the output of the trans-

former is connected to the load,  $R_{LOAD}$ , via a passive reconstruction filter or cable.  $R_{DIFF}$  is determined by the transformer's impedance ratio and provides the proper source termination that results in a low VSWR (Voltage Standing Wave Ratio). Note that approximately half the signal power will be dissipated across  $R_{DIFF}$ .

## DIFFERENTIAL COUPLING USING AN OP AMP

An op amp can also be used to perform a differential-to-single-ended conversion as shown in Figure 43. The AD9772A is configured with two equal load resistors,  $R_{LOAD}$ , of 25Ω. The differential voltage developed across  $I_{OUTA}$  and  $I_{OUTB}$  is converted to a single-ended signal via the differential op amp configuration. An optional capacitor can be installed across  $I_{OUTA}$  and  $I_{OUTB}$ , forming a real pole in a low-pass filter. The addition of this capacitor also enhances the op amp's distortion performance by preventing the DAC's high slewing output from overloading the op amp's input.

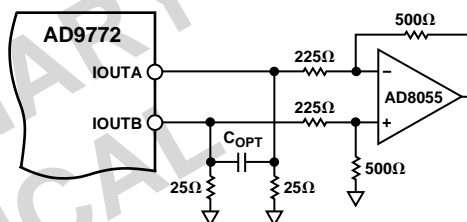


Figure 43. DC Differential Coupling Using an Op Amp

The common-mode rejection of this configuration is typically determined by the resistor matching. In this circuit, the differential op amp circuit using the AD8055 is configured to provide some additional signal gain. The op amp must operate from a dual supply since its output is approximately  $\pm 1.0$  V. A high speed amplifier, capable of preserving the differential performance of the AD9772A while meeting other system level objectives (i.e., cost, power), should be selected. The op amp's differential gain, its gain setting resistor values and full-scale output swing capabilities should all be considered when optimizing this circuit.

The differential circuit shown in Figure 44 provides the necessary level shifting required in a single supply system. In this case,  $AVDD$ , the positive analog supply for both the AD9772A and the op amp, is also used to level-shift the differential output of the AD9772A to midsupply (i.e.,  $AVDD/2$ ). The AD8057 is a suitable op amp for this application.

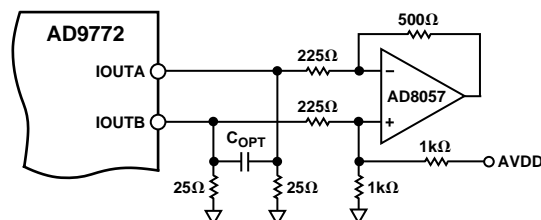


Figure 44. Single Supply DC Differential Coupled Circuit

### SINGLE-ENDED UNBUFFERED VOLTAGE OUTPUT

Figure 45 shows the AD9772A configured to provide a unipolar output range of approximately 0 V to +0.5 V for a doubly terminated 50  $\Omega$  cable since the nominal full-scale current,  $I_{OUTFS}$ , of 20 mA flows through the equivalent  $R_{LOAD}$  of 25  $\Omega$ . In this case,  $R_{LOAD}$  represents the equivalent load resistance seen by  $I_{OUTA}$ . The unused output ( $I_{OUTB}$ ) should be connected to ACOM directly. Different values of  $I_{OUTFS}$  and  $R_{LOAD}$  can be selected as long as the positive compliance range is adhered to. One additional consideration in this mode is the integral nonlinearity (INL) as discussed in the Analog Output section of this data sheet. For optimum INL performance, the single-ended, buffered voltage output configuration is suggested.

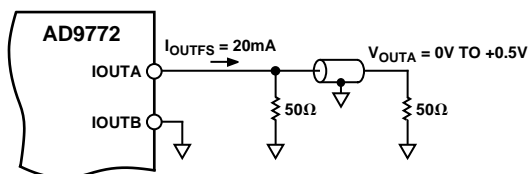


Figure 45. 0 V to +0.5 V Unbuffered Voltage Output

### SINGLE-ENDED BUFFERED VOLTAGE OUTPUT

Figure 46 shows a buffered single-ended output configuration in which the op amp U1 performs an I-V conversion on the AD9772A output current. U1 maintains  $I_{OUTA}$  (or  $I_{OUTB}$ ) at virtual ground, thus minimizing the nonlinear output impedance effect on the DAC's INL performance as discussed in the Analog Output section. Although this single-ended configuration typically provides the best dc linearity performance, its ac distortion performance at higher DAC update rates is often limited by U1's slewing capabilities. U1 provides a negative unipolar output voltage and its full-scale output voltage is simply the product of  $R_{FB}$  and  $I_{OUTFS}$ . The full-scale output should be set within U1's voltage output swing capabilities by scaling  $I_{OUTFS}$  and/or  $R_{FB}$ . An improvement in ac distortion performance may result with a reduced  $I_{OUTFS}$  since the signal current U1 will be required to sink will be subsequently reduced.

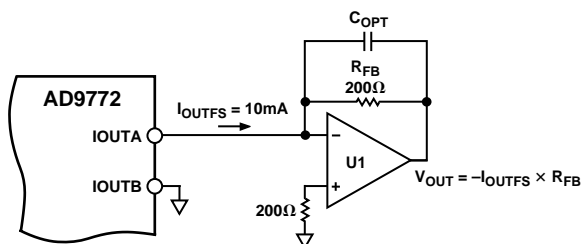


Figure 46. Unipolar Buffered Voltage Output

### POWER AND GROUNDING CONSIDERATIONS

The AD9772A contains the five following power supply inputs: AVDD, DVDD1, DVDD2, CLKVDD and PLLVDD. The AD9772A is specified to operate over a 2.8 V to 3.2 V supply range, thus accommodating +3.0 V and/or 3.3 V power supplies with up to  $\pm 10\%$  regulation. However, the following two conditions *must* be adhered to when selecting power supply sources for AVDD, DVDD1-2, CLKVDD, and PLLVDD:

1. PLLVDD = CLKVDD = 3.1V-3.5V when PLL Clock Multiplier enabled.

(Otherwise PLLVDD = PLLCOM)

2. DVDD1-2 = CLKVDD  $\pm$  0.30 V

To meet the first condition, PLLVDD must be driven by the same power source as CLKVDD with each supply input independently decoupled with a 0.1  $\mu$ F capacitor to its respective grounds. To meet the second condition, CLKVDD can share the power supply source as DVDD1-2, using the decoupling network shown in Figure 47 to isolate digital noise from the sensitive CLKVDD (and PLLVDD) supply. Alternatively, separate precision voltage regulators can be used to ensure that condition two is met.

In systems seeking to simultaneously achieve high speed and high performance, the implementation and construction of the printed circuit board design is often as important as the circuit design. Proper RF techniques must be used in device selection, placement and routing and supply bypassing and grounding. Figures 55-62 illustrate the recommended printed circuit board ground, power and signal plane layouts that are implemented on the AD9772A evaluation board.

Proper grounding and decoupling should be a primary objective in any high speed, high resolution system. The AD9772A features separate analog and digital supply and ground pins to optimize the management of analog and digital ground currents in a system. AVDD, CLKVDD, and PLLVDD must be powered from a clean analog supply and decoupled to their respective analog common (i.e., ACOM, CLKCOM and PLLCOM) as close to the chip as physically possible. Similarly, DVDD1 and DVDD2, the digital supplies, should be decoupled to DCOM.

For those applications requiring a single +3.3 V supply for both the analog, digital supply and Phase Lock Loop supply, a clean AVDD and/or CLKVDD may be generated using the circuit shown in Figure 47. The circuit consists of a differential LC filter with separate power supply and return lines. Lower noise can be attained using low ESR-type electrolytic and tantalum capacitors.

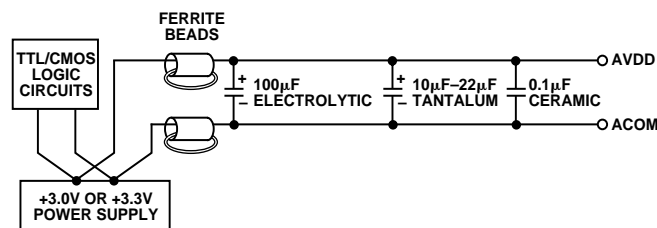


Figure 47. Differential LC Filter for +3 V or 3.3V Applications

Maintaining low noise on power supplies and ground is critical to obtain optimum results from the AD9772A. If properly implemented, ground planes can perform a host of functions on high speed circuit boards: bypassing, shielding current transport, etc. In mixed signal design, the analog and digital portions of the board should be distinct from each other, with the analog ground plane confined to the areas covering the analog signal traces,

and the digital ground plane confined to areas covering the digital interconnects.

All analog ground pins of the DAC, reference and other analog components should be tied directly to the analog ground plane. The two ground planes should be connected by a path 1/8 to 1/4 inch wide underneath or within 1/2 inch of the DAC to maintain optimum performance. Care should be taken to ensure that the ground plane is uninterrupted over crucial signal paths. On the digital side, this includes the digital input lines running to the DAC. On the analog side, this includes the DAC output signal, reference signal and the supply feeders.

The use of wide runs or planes in the routing of power lines is also recommended. This serves the dual role of providing a low series impedance power supply to the part, as well as providing some “free” capacitive decoupling to the appropriate ground plane. It is essential that care be taken in the layout of signal and power ground interconnects to avoid inducing extraneous voltage drops in the signal ground paths. It is recommended that all connections be short, direct and as physically close to the package as possible in order to minimize the sharing of conduction paths between different currents. When runs exceed an inch in length, strip line techniques with proper termination resistors should be considered. The necessity and value of this resistor will be dependent upon the logic family used.

For a more detailed discussion of the implementation and construction of high speed, mixed signal printed circuit boards, refer to Analog Devices’ application note AN-333

## APPLICATIONS MULTICARRIER

The AD9772A’s wide dynamic range performance makes it well suited for next generation base station applications in which it reconstructs multiple modulated carriers over a designated frequency band. Cellular multicarrier and multimode radios are often referred to as software radios since the carrier tuning and modulation scheme is software programmable and performed digitally. The AD9772A is the recommended TxDAC in Analog Device’s Softcell chipset which comprises the AD6622, Quadrature Digital Upconverter IC, along with its companion Rx Digital Downconverter IC, the AD6624, and 14-bit, 65 MSPS ADC, the AD6644. Figure 48 shows a generic software radio Tx signal chain based on the AD9772A/AD6622.

Figure 49 shows a spectral plot of the AD9772A operating at 64.54 MSPS reconstructing eight IS-136 modulated carriers spread over a 25 MHz band. For this particular test scenario, the AD9772A exhibited 74 dBc SFDR performance along with a carrier-to-noise ratio (CNR) of 73 dB. Figure 50 shows a spectral plot of the AD9772A operating at 52 MSPS reconstructing four equal GSM carriers spread over a 15 MHz band. The SFDR and CNR (in 100 kHz BW) measured to be 76 dBc and 83.4 dB respectively along with a channel power of -13.5 dBFS. Note, the test vectors were generated using Rohde & Schwarz’s WinIQSIM software.

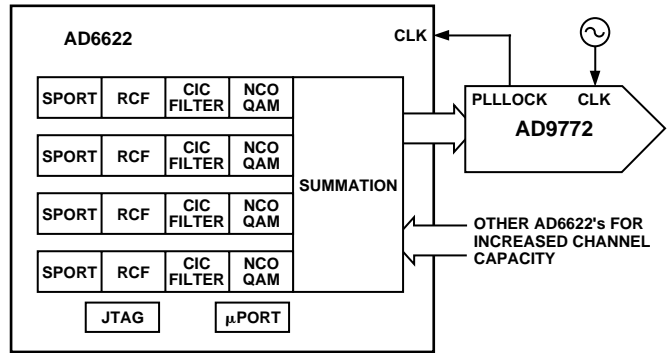


Figure 48. Generic Multicarrier Signal Chain Using the AD6622 and AD9772

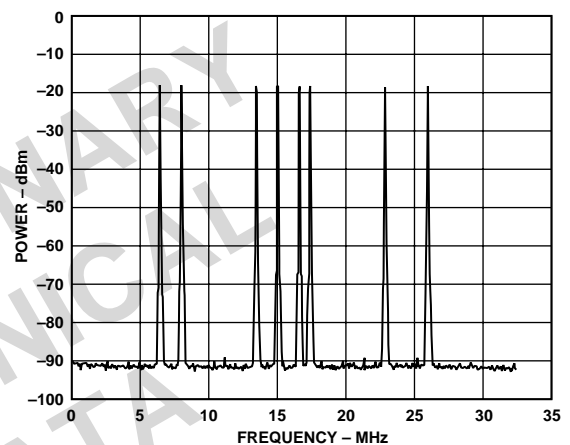


Figure 49. Spectral Plot of AD9772A Reconstructing Eight IS-136 Modulated Carriers @  $f_{DATA} = 64.54$  MSPS, PLLVDD = 0

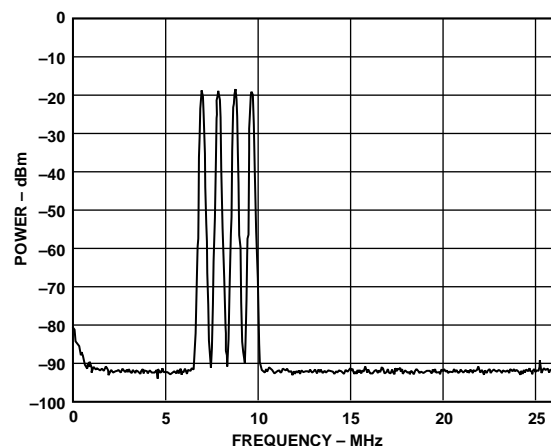


Figure 50. Spectral Plot of AD9772A Reconstructing Four GSM Modulated Carriers @  $f_{DATA} = 52$  MSPS, PLLVDD = 0

Although the above IS-136 and GSM spectral plots are representative of the AD9772A’s performance for a particular set of test conditions, the following recommendations are offered to maximize the performance and system integration of the AD9772A into multicarrier applications:

1. To achieve the highest possible CNR, the PLL Clock Multiplier should be disabled (i.e., PLLVDD to PLLCOM) and the AD9772A's clock input driven with a low jitter/phase noise clock source at twice the input data rate. In this case, the divide-by-two clock appearing at PLLLOCK should serve as the master clock for the digital upconverter IC(s) such as the AD6622. PLLLOCK should be limited to a fanout of one.
2. The AD9772A achieves its optimum noise and distortion performance when configured for baseband operation along with a differential output and a full-scale current,  $I_{OUTFS}$ , set to approximately 20 mA.
3. Although the  $2\times$  interpolation filters frequency roll-off provides a maximum reconstruction bandwidth of  $0.422 \times f_{DATA}$ , the optimum adjacent image rejection (due to the interpolation process) is achieved (i.e.,  $> 73$  dBc) if the maximum channel assignment is kept below  $0.400 \times f_{DATA}$ .
4. To simplify the subsequent IF stages filter requirements (i.e., mixer image and LO rejection), it is often advantageous to offset the frequency band from dc to relax the transition band requirements of the IF filter.
5. Oversampling the frequency band often results in improved SFDR and CNR performance. This implies that the data input rate to the AD9772A is greater than  $f_{PASSBAND}/0.4$  where  $f_{PASSBAND}$  is the maximum bandwidth in which the AD9772A will be required to reconstruct and place carriers. The improved noise performance results in a reduction in the TxDAC's noise spectral density due to the added process gain realized with oversampling. Also, higher oversampling ratios provide greater flexibility in the frequency planning.

### BASEBAND SINGLE-CARRIER

The AD9772A is also well suited for wideband single-carrier applications such as WCDMA and multilevel QAM whose modulation scheme requires wide dynamic range from the reconstruction DAC to achieve the out-of-band spectral mask as well as the in-band CNR performance. Many of these applications strategically place the carrier frequency at one quarter of the DAC's input data rate (i.e.,  $f_{DATA}/4$ ) to simplify the digital modulator design. Since this constitutes the first fixed IF frequency, the frequency tuning is accomplished at a later IF stage. To enhance the modulation accuracy as well as reduce the shape factor of the second IF SAW filter, many applications will often specify the passband of the IF SAW filter be greater than the channel bandwidth. The trade-off is that the TxDAC must now meet the particular application's spectral mask requirements within the extended passband of the 2nd IF, which may include two or more adjacent channels.

Figure 51 shows a spectral plot of the AD9772A reconstructing a test vector similar to those encountered in WCDMA applications with the following exception. WCDMA applications prescribe a root raised cosine filter with an  $\alpha = 0.22$ , which limits the theoretical ACPR of the TxDAC to about 70 dB. This particular test vector represents white noise that has been band-limited by a "brickwall" bandpass filter with the same passband such

that its maximum ACPR performance is theoretically 83 dB and its peak-to-rms ratio is 12.4 dB. As Figure 51 reveals, the AD9772A is capable of approximately 78 dB ACPR performance when one accounts for the additive noise/distortion contributed by the FSEA30 spectrum analyzer.

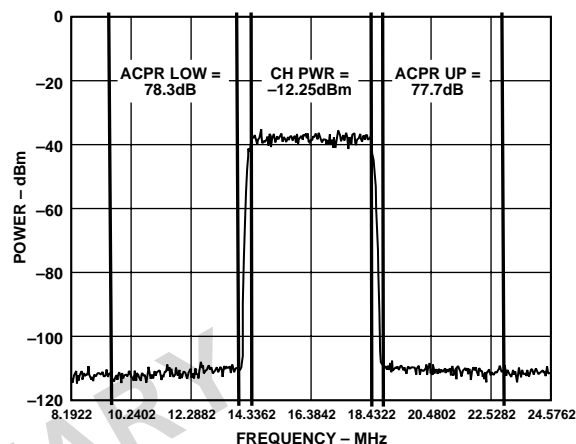


Figure 51. AD9772A Achieves 78 dB ACPR Performance Reconstructing a "WCDMA-Like" Test Vector with  $f_{DATA} = 65.536$  MSPS and PLLVDD = 0

### DIRECT IF

As discussed in the Digital Modes of Operation section, the AD9772A can be configured to transform digital data representing baseband signals into IF signals appearing at odd multiples of the input data rate (i.e.,  $N \cdot f_{DATA}$  where  $N = 1, 3, \dots$ ). This is accomplished by configuring the MOD1 and MOD0 digital inputs HIGH. Note, the maximum DAC update rate of 400 MSPS limits the data input rate in this mode to 100 MSPS when the "zero-stuffing operation" is enabled (i.e., MOD1 High). Applications requiring higher IFs (i.e., 140 MHz) using higher data rates should disable the "zeros-stuffing" operation. Also, to minimize the effects of the PLL Clock Multipliers phase noise as shown in Figure 27, an external low jitter/phase noise clock source equal to  $4 \times f_{DATA}$  is recommended.

Figure 52 shows the actual output spectrum of the AD9772A reconstructing a 16-QAM test vector with a symbol rate of 5 MSPS. The particular test vector was centered at  $f_{DATA}/4$  with  $f_{DATA} = 100$  MSPS, and  $f_{DAC} = 400$  MHz. For many applications, the pair of images appearing around  $f_{DATA}$  will be more attractive since they have the flattest passband and highest signal power. Higher images can also be used with the understanding that these images will have reduced passband flatness, dynamic range, and signal power, thus reducing the CNR and ACP performance. Figure 53 shows a dual tone SFDR amplitude sweep at the various IF images with  $f_{DATA} = 100$  MSPS and  $f_{DAC} = 400$  MHz and the two tones centered around  $f_{DATA}/4$ . Note, since an IF filter is assumed to precede the AD9772A, the SFDR was measured over a 25 MHz window around the images occurring at 75 MHz, 125 MHz, 275 MHz and 325 MHz.

Regardless of what image is selected for a given application, the adjacent images must be sufficiently filtered. In

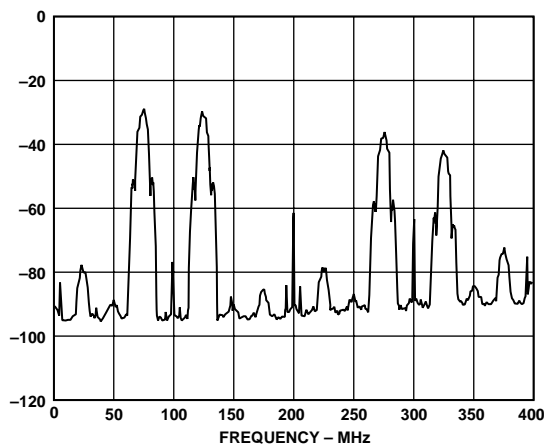


Figure 52. Spectral Plot of 16-QAM Signal in Direct IF Mode at  $f_{DATA} = 100$  MSPS

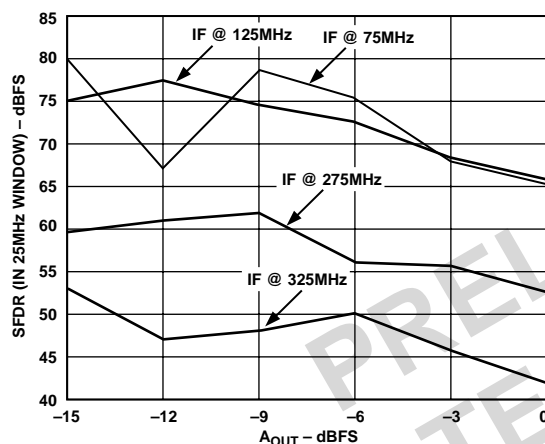


Figure 53. Dual Tone "Windowed" SFDR vs.  $A_{OUT}$  at  $f_{DATA} = 100$  MSPS

most cases, a SAW filter providing differential inputs represents the optimum device for this purpose. For single-ended SAW filters, a balanced-to-unbalanced RF transformer is recommended. The AD9772A's high output impedance provides a certain amount of flexibility in selecting the optimum resistive load,  $R_{LOAD}$ , as well as any matching network.

For many applications, the data update rate to the DAC (i.e.,  $f_{DATA}$ ) must be some fixed integer multiple of some system reference clock (i.e.,  $GSM = 13$  MHz). Furthermore, these applications prefer to use standard IF frequencies which offer a large selection of SAW filter choices of varying passbands (i.e., 70 MHz). These applications may still benefit from the AD9772A's direct IF mode capabilities when used in conjunction with a digital upconverter such as the AD6622. Since the AD6622 can digitally synthesize and tune up to four modulated carriers, it is possible to judiciously tune these carriers in a region which may fall within an IF filter's passband upon reconstruction by the AD9772A. Figure 54 shows an example in which four carriers were tuned around 18 MHz with a digital upconverter operating at 52 MSPS such that when reconstructed by the AD9772A in the IF MODE, these carriers fall around a 70 MHz IF.

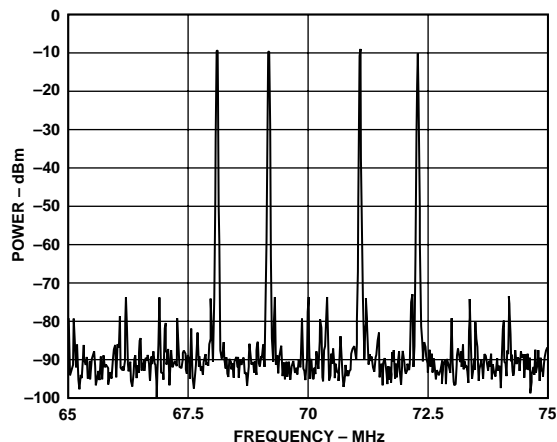


Figure 54. Spectral Plot of Four Carriers at 70MHz IF with  $F_{data} = 52$  MSPS,  $PLLVDD = 0$

#### AD9772A EVALUATION BOARD

The AD9772-EB is an evaluation board for the AD9772A TxDAC. Careful attention to layout and circuit design, combined with prototyping area, allows the user to easily and effectively evaluate the AD9772A in different modes of operation.

Referring to Figures 55 and 56, the AD9772A's performance can be evaluated differentially or single-ended using a transformer, differential amplifier, or directly coupled output. To evaluate the output differentially using the transformer, remove jumpers JP12 and JP13 and monitor the output at J6 (IOUT). To evaluate the output differentially, remove the transformer (T2) and install jumpers JP12 and JP13. The output of the amplifier can be evaluated at J13 (AMPOUT). To evaluate the AD9772A single-ended and directly coupled, remove the transformer and jumpers (JP12 and JP13) and install resistors R16 or R17 with 0  $\Omega$ .

The digital data to the AD9772A comes across a ribbon cable which interfaces to a 40-pin IDC connector. Proper termination or voltage scaling can be accomplished by installing RN2 and/or RN3 SIP resistor networks. The 22  $\Omega$  DIP resistor network, RN1, must be installed and helps reduce the digital data edge rates. A single-ended CLOCK input can be supplied via the ribbon cable by installing JP8 or more preferably via the SMA connector, J3 (CLOCK). If the CLOCK is supplied by J3, the AD9772A can be configured for a differential clock interface by installing jumpers JP1 and configuring JP2, JP3, and JP9 for the DF position. To configure the AD9772A clock input for a single-ended clock interface, remove JP1 and configure JP2, JP3 and JP9 for the SE position.

The AD9772A's PLL clock multiplier can be disabled by configuring jumper JP5 for the L position. In this case, the user must supply a clock input at twice ( $2\times$ ) the data rate via J3 (CLOCK). The  $1\times$  clock is made available on SMA connector J1 (PLLLOCK), and should be used to trigger a pattern generator directly or via a programmable pulse generator. Note, PLLLOCK is capable of providing a 0 V to 0.85 V output into a 50  $\Omega$  load. To enable the PLL clock multiplier, JP5 must be configured for the H position. In this case, the clock may be supplied via the ribbon cable (i.e., JP8 installed) or J3 (CLOCK).



The divide-by-N ratio can be set by configuring JP6 (DIV0) and JP7 (DIV1).

The AD9772A can be configured for Baseband or Direct IF Mode operation by configuring jumpers JP11 (MOD0) and JP10 (MOD1). For baseband operation, JP10 and JP11 should be configured in the L position. For direct IF operation, JP10 and JP11 should be configured in the H position. For direct IF operation without “zero-stuffing,” JP11 should be configured in the H position while JP10 should be configured in the low position.

The AD9772A's voltage reference can be enabled or disabled via JP4 (EXT REF IN). To enable the reference, configure JP in the INT position. A voltage of approximately 1.2 V will appear at the TP6 (REFIO) test point. To disable the internal reference, configure JP4 in the EXT position and drive TP6 with an external voltage reference. Lastly, the AD9772A can be placed in the SLEEP mode by driving the TP11 test point with logic level HIGH input signal.

PRELIMINARY  
TECHNICAL  
DATA

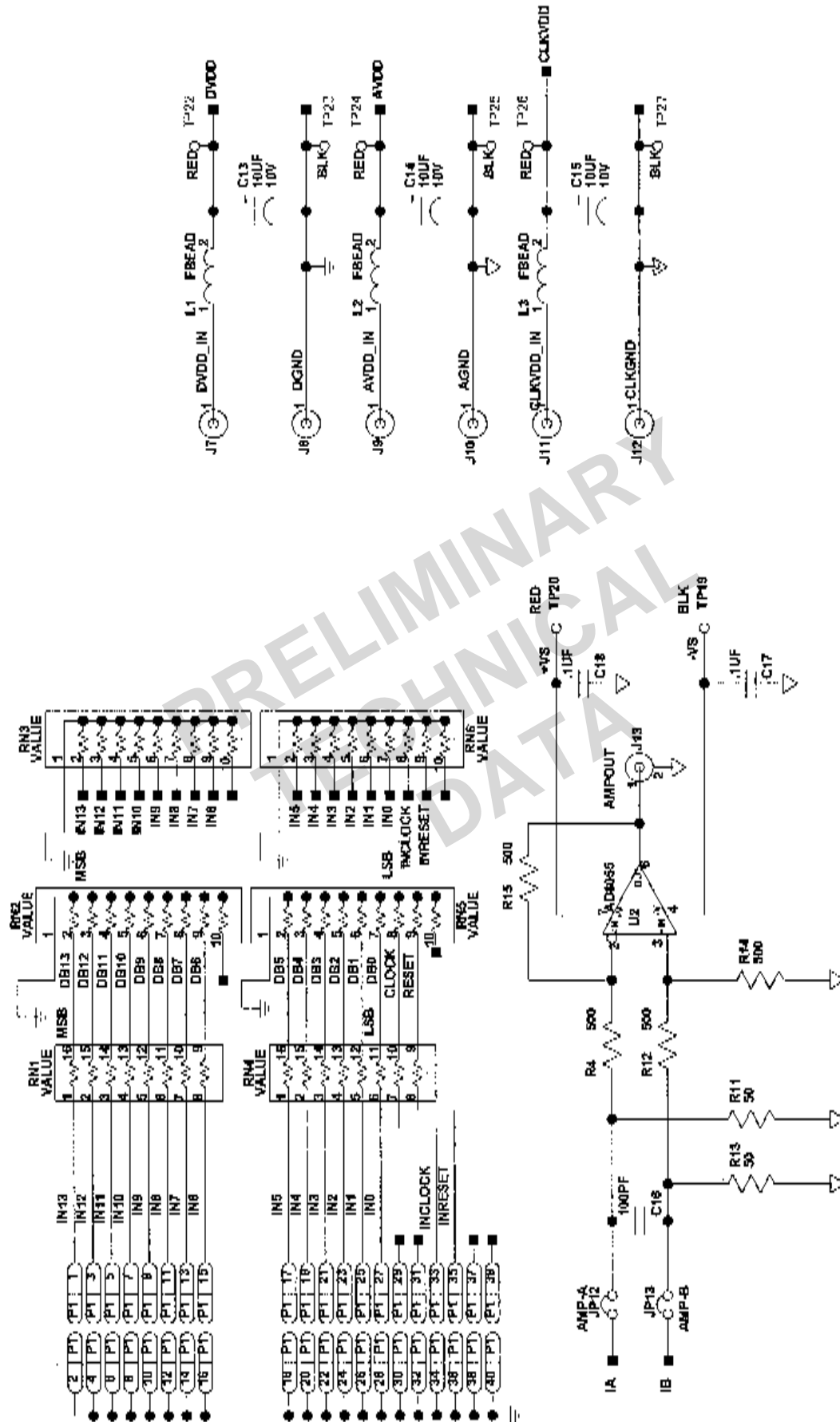


Figure 55. Drafting Schematic of Evaluation Board

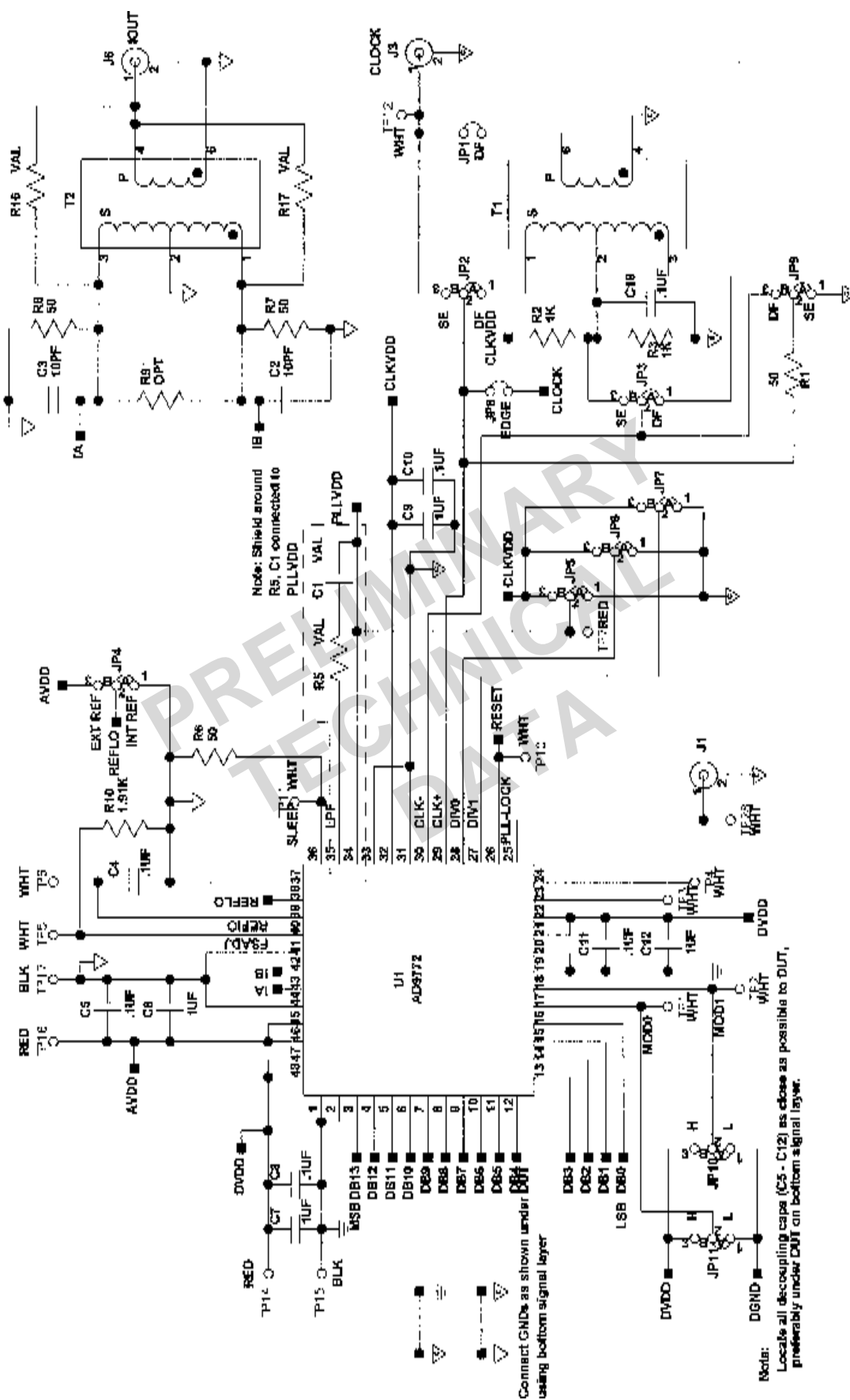


Figure 56. Drafting Schematic of Evaluation Board (Continued)

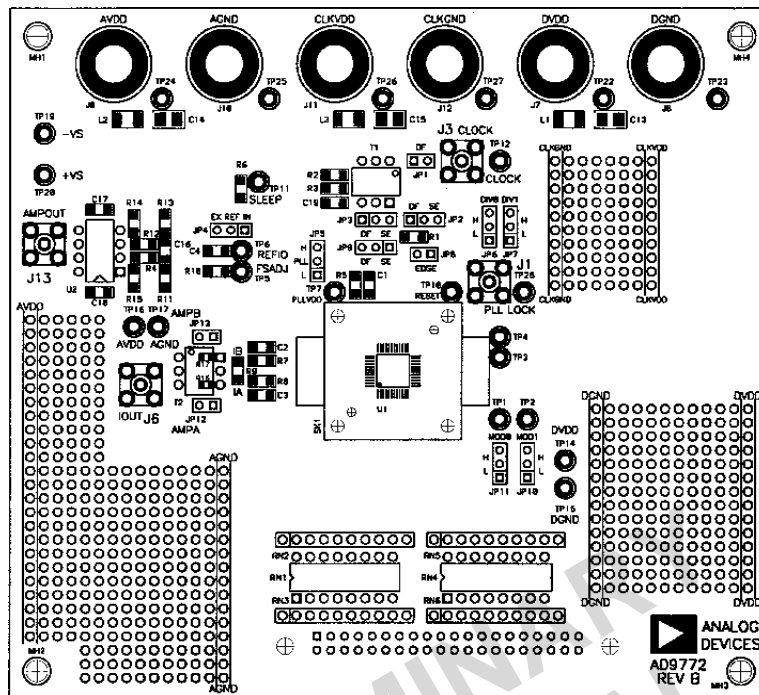


Figure 57. Silkscreen Layer-Top

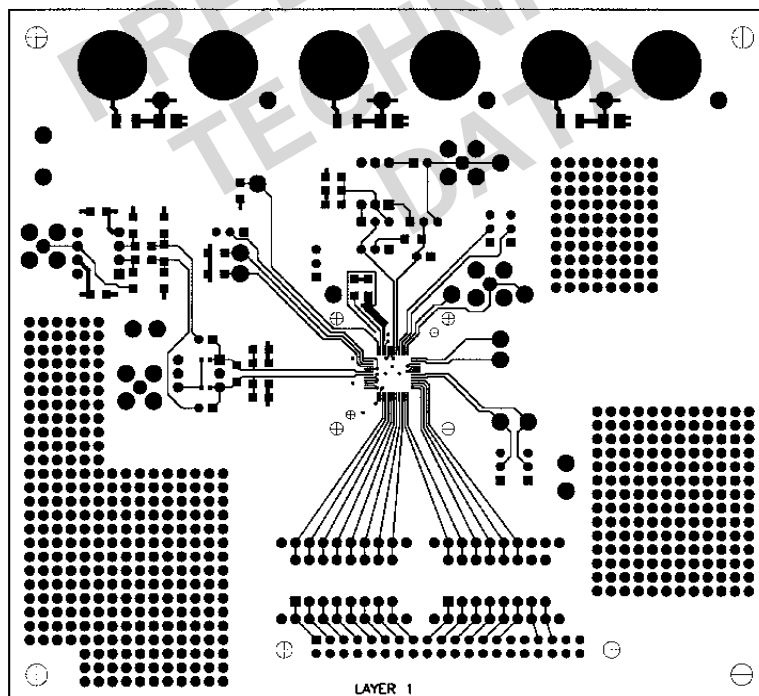


Figure 58. Component Side PCB Layout (Layer 1)

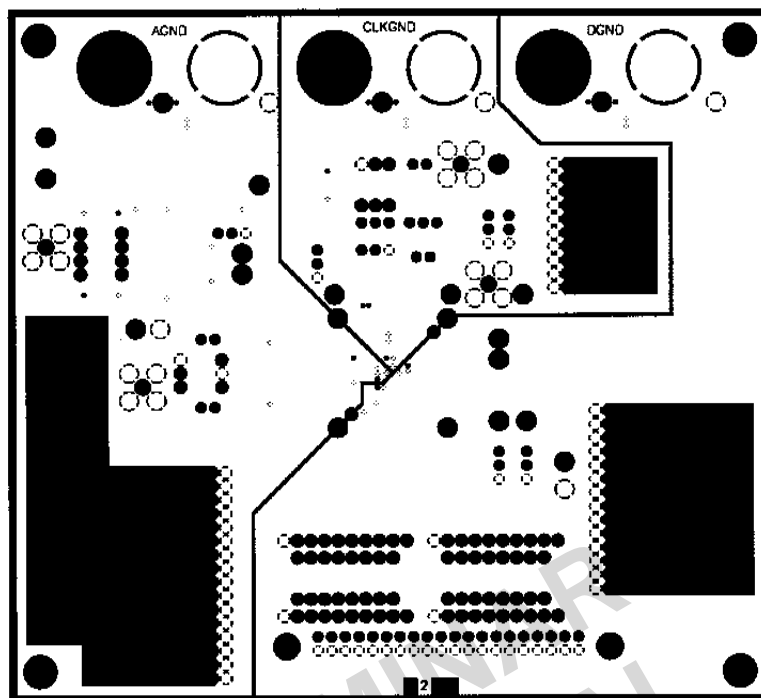


Figure 59. Ground Plane PCB Layout (Layer 2)

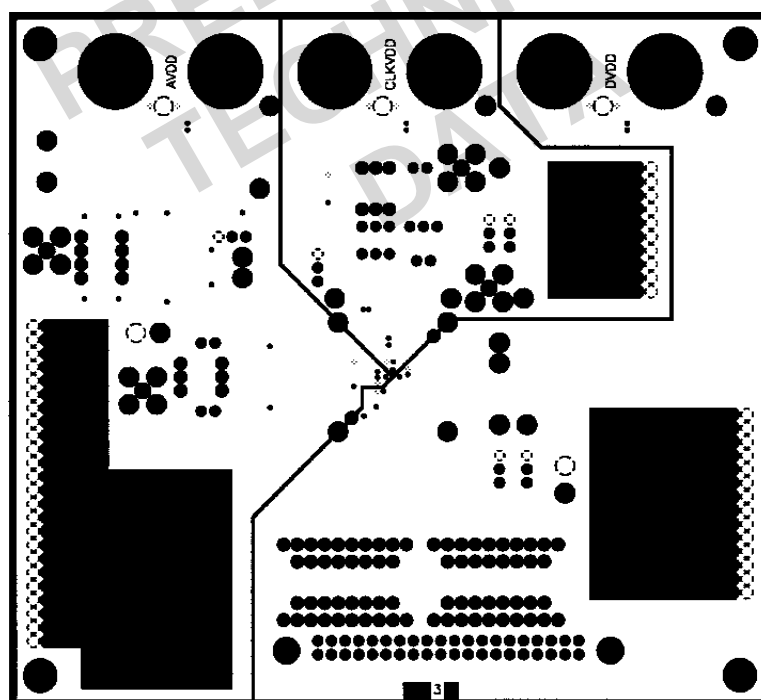


Figure 60. Power Plane PCB Layout (Layer 3)

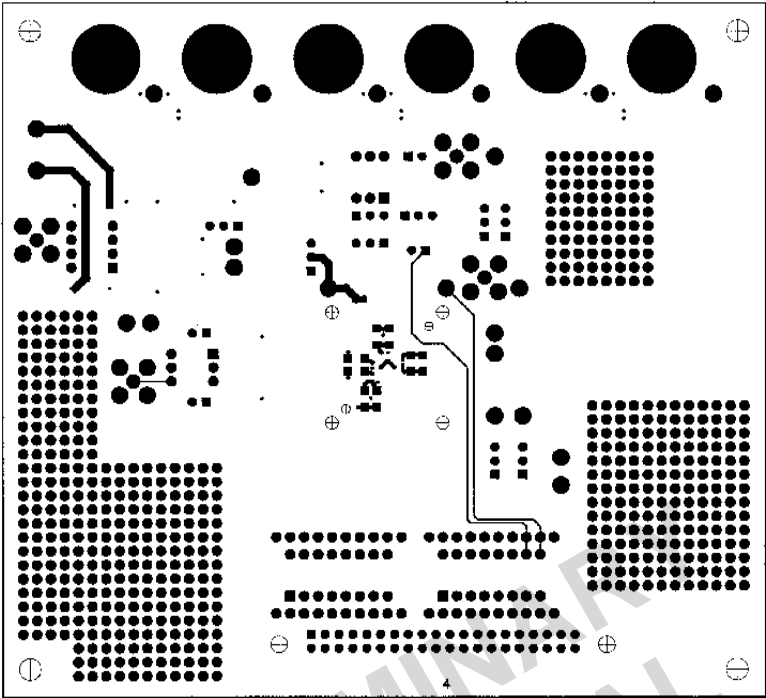


Figure 61. Solder Side PCB Layout (Layer 4)

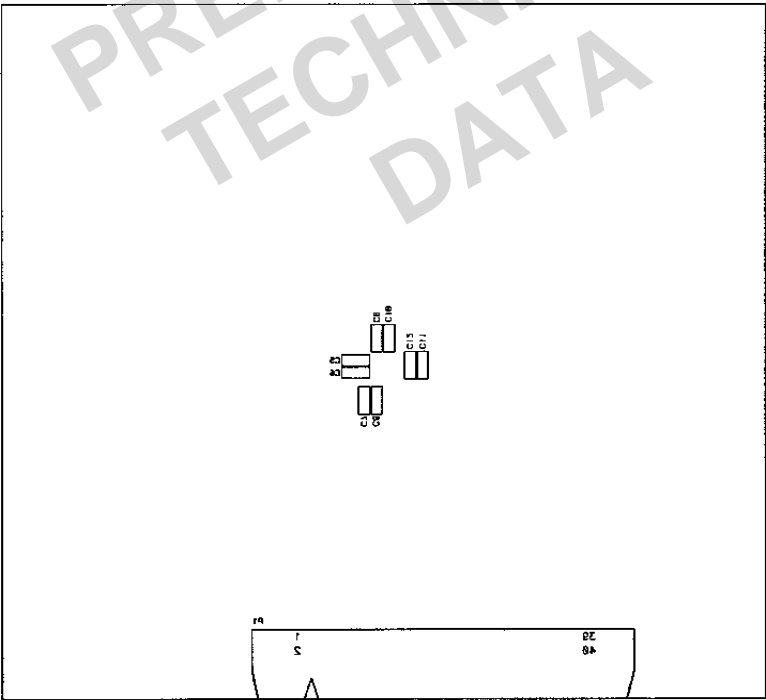


Figure 62. Silkscreen Layer-Bottom

OUTLINE DIMENSIONS  
Dimensions shown in inches and (mm).

48-Lead Thin Plastic Quad Flatpack  
(ST-48)

