# Audio Precision System Two Cascade *Plus*

# Specifications

(Effective with APWIN version 2.1 Software)





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# Specifications

# **Analog Signal Outputs**

All System Two Cascade *Plus* configurations, except the SYS-2700, contain an ultra-low distortion analog sine wave generator and two independent transformer-coupled output stages.

The SYS-2622 and SYS-2722 configurations also contain a dual-channel D/A signal generator for enhanced capabilities. Option "BUR" adds analog-generated sine burst, square wave, and noise signals. Option "IMD" adds analog-generated IMD test signals.

Unless otherwise noted, all specifications are valid only for outputs  $\geq 150 \ \mu Vrms$  [420  $\mu Vpp$ ].

# **Analog Output Characteristics**

Source Configuration	Selectable balanced, unbalanced, or CMTST (common mode test)
Source Impedances Balanced or CMTST Unbalanced	40 Ω (±1 Ω), 150 Ω <sup>1</sup> (±1.5 Ω), or 600 Ω (±3 Ω) 20 Ω (±1 Ω) or 600 Ω (±3 Ω)
Max Floating Voltage	42 Vpk (outputs are isolated from each other)
Output Current Limit	Typically >80 mA
Max Output Power into 600 Ω Balanced Unbalanced	+30.1 dBm (Rs = 40 Ω) +24.4 dBm (Rs = 20 Ω)
Output Related Crosstalk 10 Hz–20 kHz 20 kHz–100 kHz	$\leq$ –120 dB or 5 $\mu$ V, whichever is greater $\leq$ –106 dB or 10 $\mu$ V, whichever is greater

# **Low Distortion Sine Wave Generator**

Frequency Range	10 Hz–204 kHz
Frequency Accuracy	
High-accuracy mode	±0.03%
Fast mode	±0.5%

 $^1$  200  $\Omega$   $\pm 2$   $\Omega$  with option "EURZ"

0.005% 0.025 Hz, 10 Hz–204.75 Hz;
0.025 Hz, 10 Hz–204.75 Hz;
0.25 Hz, 205 Hz–2.0475 kHz;
2.5 Hz, 2.05 kHz–20.475 kHz;
25 Hz, 20.5 kHz–204 kHz
<10 µV to 26.66 Vrms [+30.7 dBu]
<10 µV to 13.33 Vrms [+24.7 dBu]
±0.7% [±0.06 dB] at 1 kHz
0.003 dB or 0.05 µVrms, whichever is larger
±0.008 dB (typically <0.003 dB)
±0.03 dB
±0.10 dB
+0.2 / –0.3 dB
≤(0.00025% + 1.0 μV) [–112 dB], 22 kHz BW (valid only for analyzer inputs ≤8.5 Vrms)
≤(0.0003% + 1.0 μV) [–110.5 dB], 22 kHz BW
≤(0.0005% + 2.0 µV) [–106 dB], 80 kHz BW
≤(0.0010% + 5.0 µV) [–100 dB], 500 kHz BW
≤(0.0040% + 5.0 µV) [–88 dB], 500 kHz BW

#### **Intermodulation Distortion Test Signals**

with option "IMD"

#### SMPTE (or DIN)

LF Tone	40, 50, 60, 70, 100, 125, 250, or 500 Hz; all ±1.5%
HF Tone Range	2 kHz–200 kHz
Mix Ratio	4:1 or 1:1 (LF:HF)
Amplitude Range <sup>5</sup>	
Balanced	30 μVpp to 75.4 Vpp
Unbalanced	30 µVpp to 37.7 Vpp
Amplitude Accuracy	±2.0% [±0.17 dB]
Residual IMD <sup>6</sup>	0.0015% [–96.5 dB], 60+7 kHz or 250+8 kHz

<sup>&</sup>lt;sup>2</sup> 20 Hz–50 kHz only. Decrease maximum output by a factor of 2 (–6.02 dB) for the full 10 Hz–204 kHz range.

<sup>&</sup>lt;sup>3</sup> System specification measured with the Cascade Plus analog analyzer set to the indicated measurement bandwidth (BW). Generator amplitude setting must be ≤12 Vrms balanced or ≤6 Vrms unbalanced for specified performance below 30 Hz. At higher amplitude settings generator THD derates to 0.0020% from 20 Hz–30 Hz.

<sup>&</sup>lt;sup>4</sup> Individual harmonics are typically <-130 dBc at 1 kHz, and <-120 dBc from 25 Hz to 20 kHz measured with a passive notch filter and FFT analyzer.

<sup>&</sup>lt;sup>5</sup> Calibration with other amplitude units is based upon an equivalent sinewave having the same Vpp amplitude.

 $<sup>^{6}</sup>$  System specification measured with the Cascade Plus analog analyzer at any amplitude  $\geq$ 200 mVrms.

#### CCIF and DFD

Difference Frequency	80, 100, 120, 140, 200, 250, 500 or 1 kHz; all ±1.5%
Center Frequency	4.5 kHz–200 kHz
Amplitude Range⁵	
Balanced	30 µVpp to 75.4 Vpp
Unbalanced	30 µVpp to 37.7 Vpp
Amplitude Accuracy	±3.0% [±0.26 dB]
CCIF Residual IMD <sup>6</sup>	≤0.0004% [–108 dB], 14 kHz+15 kHz (odd order & spurious typ <0.05%)
DFD Residual IMD <sup>6</sup>	≤0.0002% [–114 dB], 14 kHz+15 kHz (odd order & spurious typ <0.025%)

#### DIM (or TIM)

Squarewave Frequency	3.15 kHz (DIM-30 and DIM-100); 2.96 kHz (DIM-B); both ±1%
Sinewave Frequency	15 kHz (DIM-30 and DIM-100); 14 kHz (DIM-B)
Amplitude Range <sup>4</sup> Balanced Unbalanced	30 µVpp to 75.4 Vpp 30 µVpp to 37.7 Vpp
Amplitude Accuracy	±2.0% [±0.17 dB]
Residual IMD <sup>5</sup>	≤0.0020% [–94 dB]

# **Special Purpose Signals**

with option "BUR"

# Sine Burst

Sinc Durst	
Frequency Range	20 Hz–100 kHz
Frequency Accuracy	Same as Sinewave
ON Amplitude Range Accuracy, Flatness	Same as Sinewave
OFF Ratio Range	0 dB to80 dB
OFF Ratio Accuracy	±0.3 dB, 0 to –60 dB
ON Duration	1 to 65535 cycles, or externally gated
Interval Range	2 to 65536 cycles

#### Square Wave

Frequency Range	20 Hz–20 kHz
Frequency Accuracy	Same as Sinewave
Amplitude Range <sup>4</sup> Balanced Unbalanced	30 μVpp to 37.7 Vpp 30 μVpp to 18.8 Vpp
Amplitude Accuracy	±2.0% [±0.17 dB] at 400 Hz
Rise/fall time	Typically 2.0 μs

#### Noise Signals

Bandwidth limited 10 Hz-23 kHz
Bandwidth limited 20 Hz–200 kHz
Approximately 1/3-octave (2-pole) filtered pink noise, continuously tunable from 20 Hz–100 kHz
True random or pseudo-random
Typically 262 ms (synchronized to the analyzer 4/s reading rate)
(Approximate calibration only)
30 μVpp to 37.7 Vpp
30 µVpp to 18.8 Vpp

# **Graphs of Typical Analog Generator Performance**

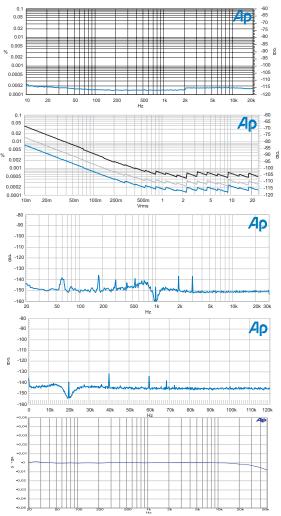


Figure 1. Typical system THD+N versus Frequency at 2 Vrms (analog sine)

Figure 2. Typical system THD+N versus amplitude at 1 kHz. Lower trace is with 22 kHz bandwidth limiting. Middle trace is with 80 kHz. Upper trace is with 500 kHz.

Figure 3. Typical residual THD+N spectrum at 1 kHz, 2 Vrms. (32768 point FFT of notch filter output, SR = 65.536 ks/s, 16 averages).

Figure 4. Typical residual THD+N spectrum at 20 kHz, 2 Vrms. (32768 point FFT of notch filter output, SR = 262 ks/s, 16 averages).

Figure 5. Typical analog system flatness at 2 Vrms signal level (measured with the analog analyzer's Level meter, dc input coupling)

# **D/A Generated Analog Signals**

Available only in the SYS-2622 and SYS-2722 configurations. Except for arbitrary waveforms, the digitally-generated analog signals and the digital output signals are independently selectable and concurrently available. If both analog and digital outputs are selecting Arbitrary Waveform, it must be the same one.

# **Common Specifications**

Sample Rate (SR) Sine, IMD signals Other signals	fixed at 65.536 ks/s or 131.072 ks/s 7.2 ks/s to 108 ks/s variable; or fixed at 65.536 ks/s or 131.072 ks/s
Frequency Accuracy	$\pm 0.0002\%$ [2 PPM] internal reference, lockable to external reference
D/A Resolution	24-bit sigma-delta

# "SINE (D/A)" Signal Family

The Sine family includes "Normal," "Var Phase," "Stereo," "Dual," "Shaped Burst," and "EQ Sine." Normal and EQ Sine produce a monaural signal with the best (lowest) residual THD+N performance. EQ Sine varies the amplitude in accordance with a selected EQ file. Var Phase produces the same sine wave in both channels but with settable phase offset. Stereo provides sine waves of independently settable frequency in each channel (phase is random if both frequencies are set equal). Dual produces a monaural test signal containing a mixture of two sine waves of independently settable frequencies are monaural sine burst signal with a raised cosine amplitude envelope (see Figure 30 on page 25).

Frequency Ranges	10 Hz to 30 kHz (65.536 ks/s), or 10 Hz to 60 kHz (131.072 ks/s)
Frequency Resolution	Sample Rate ÷ 2 <sup>23</sup> [0.0078 Hz in the 30 kHz range]
Flatness (1 kHz ref)	
20 Hz–20 kHz ´	±0.01 dB
10 Hz–30 kHz	±0.03 dB
30 kHz–50 kHz	±0.10 dB (typically –0.5 dB at 60 kHz)
THD+N <sup>7</sup> (20Hz–20kHz)	
30 kHz range	0.0007% [–103 dB]
60 kHz range	0.0014% [–97 dB]
Variable Phase Range	–180.0 to +179.9 deg
Dual-Sine Ratio Range	0 dB to –100 dB, usable to –138 dB
Shaped Burst Interval	2 to 65536 cycles
Shaped Burst On Time	1 to (number of interval cycles minus 1)

<sup>&</sup>lt;sup>7</sup> System specification measured with the Cascade Plus analog analyzer set for a 22 kHz measurement BW.

# "IMD (D/A)" Signal Family

#### SMPTE/DIN Test Signal

LF Tone	40 Hz to 500 Hz
HF Tone	2.00 kHz to 50 kHz
Mix Ratio	4:1 or 1:1 (LF:HF)
Residual IMD <sup>7</sup>	≤0.0010% [–100 dB], 60/7 kHz or 250/8 kHz

#### CCIF/DFD Test Signal

Difference Frequency	80 Hz to 2 kHz
Center Frequency	4.50 kHz to >50 kHz
Residual CCIF IMD <sup>8</sup>	CCIF: ≤0.0004% [–108 dB], 14 kHz/15 kHz DFD: ≤0.0002% [–114 dB], 14 kHz/15 kHz

#### DIM Test Signal

Squarewave Frequency	3.15 kHz for DIM30 and DIM100; 2.96 kHz for DIMB
Sinewave Frequency	15.00 kHz for DIM30 and DIM100, 14.00 kHz for DIMB
Residual IMD <sup>8</sup>	≤0.0020% [–94 dB]

# **Other Signals**

#### Arbitrary Waveform and Multitone ("Arb Wfm")

Signal	Determined by specified file name
Length	256 to 16384 points per channel. Utility is provided to prepare waveform from frequency, amplitude, and phase data.
Frequency Range	20 Hz to 47% of sample rate
Frequency Resolution	Sample rate ÷ Length [2.93 Hz at 48 ks/s and 16384 Length]
Maximum Number of Tones	(Length / 2) minus1 [8191 for Length = 16384]

#### Maximum Length Sequence ("MLS")

Sequences	Four pink, four white
Sequence Length	"32k" (32767) or "128k" (131071)
Frequency Range	10 Hz to 43% of sample rate, ±0.1 dB

#### **Special Signals**

Polarity	Sum of two sine waves phased for reinforcement with normal polarity.
Pass Thru	Passes the embedded audio signal from the rear panel Reference Input. Ratio of reference rate to output Sample Rate may not exceed 8:1.

 $<sup>^8\,</sup>$  System specification measured with the Cascade Plus analog analyzer at any voltage  ${\geq}200\,$  mVrms.

#### **Squarewave**

Frequency Range	20 Hz–20.0 kHz	
Risetime	Typically 2.0 μs	

# Noise Signal

True random white

# **Analog Analyzer**

All System Two Cascade Plus configurations, except SYS-2700, contain an input module with two independent auto-ranging input stages, each having its own level (rms) and frequency meters; a phase meter; plus a single channel multi-function analyzer module providing additional signal processing and gain stages. Standard analog analyzer functions include amplitude and noise (both wideband and selective), THD+N, and crosstalk.

The SYS-2622 and SYS-2722 configurations add dual-channel A/D converters for FFT and other special forms of analysis. Option "IMD" adds inter-modulation distortion measurement capability. Option "W&F" adds wow & flutter measurement capability.

Unless otherwise noted, all specifications assume dc coupling, rms detection, and auto-ranging operation.

Input Ranges	40 mV to 160 V in 6.02 dB steps
Maximum Rated Input	230 Vpk, 160 Vrms (dc to 20 kHz); overload protected in all ranges
Input Impedance	
Balanced	200 k $\Omega$ / 95 pF (differential)
Unbalanced	100 kΩ /185 pF
Terminations	Selectable 600 $\Omega$ or 300 $\Omega,$ each $\pm$ 1%; 1 Watt [+30 dBm] maximum power
CMRR <sup>9</sup>	
40 mV–2.5 V ranges	≥80 dB, 10 Hz–20 kHz
5 V and 10 V ranges	≥65 dB, 10 Hz–20 kHz
20 V–160 V ranges	≥50 dB, 10 Hz–1 kHz
Input Related Crosstalk	
10 Hz–20 kHz	$\leq$ –140 dB or 1 $\mu$ V, whichever is greater
20 kHz–100 kHz	$\leq$ –126 dB or 2.5 $\mu V,$ whichever is greater

#### **Analog Input Characteristics**

# Level Meter Related

Measurement Range	5 mV to 160 V for specified accuracy and flatness, usable to <100 $\mu V$
Resolution (full scale) <sup>10</sup>	
4/s and 8/s	1/40,000 [0.00022 dB]
16/s	1/20,000 [0.00043 dB]
32/s	1/10,000 [0.00087 dB]
64/s	1/5,000 [0.0017 dB]
128/s	1/2,500 [0.0035 dB]
Accuracy (1 kHz)	±0.5% [±0.05 dB]

<sup>&</sup>lt;sup>9</sup> Not valid below 50 Hz with ac coupling.

<sup>&</sup>lt;sup>10</sup> Resolution within a given range is equal to its full scale value multiplied by the fraction indicated for the selected reading rate. (Example: 40 mV input range reading resolution = 4 μV, using the 32/s reading rate). Numerical displays using a dB unit are rounded to the nearest 0.001 dB.

Flatness (1 kHz ref) <sup>11</sup>	
20 Hz–20 kHz	±0.008 dB (typically <0.003 dB)
15 Hz–50 kHz	±0.03 dB
10 Hz–120 kHz	±0.10 dB
120 kHz–200 kHz	+0.2 / –0.3 dB (typically <–0.5 dB at 500 kHz)

#### Frequency Meter Related

Measurement Range	10 Hz–500 kHz
Accuracy	±0.0006% [±6 PPM]
Resolution	6 digits + 0.000244 Hz
Minimum Input	5 mV

#### **Phase Measurement Related**

Measurement Ranges	±180, –90 / +270, or 0 / +360 deg
Accuracy <sup>12</sup>	
10 Hz–5 kHz	±0.5 deg
5 kHz–20 kHz	±1 deg
20 kHz–50 kHz	±2 deg
Resolution	0.1 deg
Minimum Input	5 mV, both inputs

# Wideband Amplitude/Noise Function

Measurement Range	<1 µV to 160 Vrms
Accuracy (1 kHz)	±1.0% [±0.09 dB]
Flatness (1 kHz ref) <sup>11</sup>	
20 Hz–20 kHz	±0.02 dB
15 Hz–50 kHz	±0.05 dB
50 kHz–120 kHz	±0.15 dB
120 kHz–200 kHz	+0.2 dB / –0.3 dB (typically < –3 dB at 500 kHz)
Bandwidth Limiting Filters	see Figure 6
LF –3 dB	<10 Hz,

<10 Hz, 22 Hz per IEC468 (CCIR), 100 Hz ±5% (3-pole), or 400 Hz ±5% (3-pole)

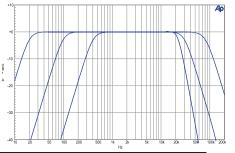


Figure 6. Typical responses of the standard band-limiting filters.

- $^{11}\textsc{Derate}$  flatness above 5 kHz by an additional  $\pm 0.02$  dB in the 20 V, 40 V, 80 V, and 160 V input ranges.
- $^{12}$  Both analyzer input channels must have same coupling (ac or dc) selection. Accuracy is valid for any input signal amplitude ratio up to  $\pm 30$  dB.

HF –3 dB	22 kHz per IEC468 (CCIR), 30 kHz ±5% (3-pole), 80 kHz ±5% (3-pole), or >500 kHz
Optional Filters	up to 7 (see section on Option Filters)
Detection	RMS ( $\tau$ = 25 ms or 50 ms), Average, QPk per IEC468 (CCIR), Pk (pseudo-peak), or S-Pk (0.7071 × Pk reading)
Residual Noise	
22 Hz–22 kHz BW	≤1.0 μV [–117.8 dBu]
80 kHz BW	≤2.0 μV [–111.8 dBu]
500 kHz BW	≤6.0 μV [–103.8 dBu]
A-weighted	≤0.5 μV [–123.8 dBu]
CCIR-QPk	≤2.5 μV [–109.8 dBu]

#### **Bandpass Amplitude Function**

Tuning Range (f <sub>o</sub> )	10 Hz–200 kHz
Tuning Accuracy	±2%
Bandpass Response	1/3-octave class II (4-pole); < –32 dB at 0.5 f_ and 2.0 f_
Accuracy (at f <sub>o</sub> )	±0.3 dB, 20 Hz–120 kHz
Residual Noise	
10 Hz–5 kHz	≤0.25 μV [–130 dBu]
5 kHz–20 kHz	≤0.5 μV [–124 dBu]
20 kHz–200 kHz	≤1.5 μV [–114 dBu]

#### **Bandreject Amplitude Function**

Tuning Range (f <sub>o</sub> )	10 Hz–200 kHz
Tuning Accuracy	±2%
Bandreject Response	typically –3 dB at 0.73 f <sub>o</sub> & 1.37 f <sub>o</sub> –20 dB at f <sub>o</sub> ±10% –40 dB at f <sub>o</sub> ±2.5%
Accuracy	$\pm 0.3$ dB, 20 Hz–120 kHz (excluding 0.5 f_o to 2.0 f_o)

#### THD+N Function

Fundamental Range	10 Hz–200 kHz
Measurement Range	0–100%
Accuracy	±0.3 dB, 20 Hz–120 kHz harmonics
Measurement Bandwidth	
LF –3 dB	<10, 22, 100, or 400 Hz
HF –3 dB	22k, 30k, 80k, or >500 kHz
	(Option filter selection also affects bandwidth)
Residual THD+N <sup>13</sup>	
At 1 kHz	≤(0.00025% + 1.0 μV) [–112 dB], 22 kHz BW

<sup>13</sup> System specification measured with the Cascade Plus analog generator and the analog analyzer set to the indicated measurement bandwidth (BW). Generator amplitude setting must be ≤12 Vrms balanced or ≤6 Vrms unbalanced for specified system performance below 30 Hz. At higher amplitude settings generator THD derates to 0.0020% from 20 Hz–30 Hz.

20 Hz–20 kHz 10 Hz–100 kHz	(valid only for analyzer inputs ≤8.5 Vrms.) ≤(0.0003% + 1.0 $\mu$ V) [–110.5 dB], 22 kHz BW ≤(0.0005% + 2.0 $\mu$ V) [–106 dB], 80 kHz BW ≤(0.0010% + 5.0 $\mu$ V) [–100 dB], 500 kHz BW ≤(0.0040% + 5.0 $\mu$ V) [–88 dB], 500 kHz BW
Minimum Input	5 mV for specified accuracy, usable to <100 $\mu V$ with fixed notch tuning
Notch Tuning Modes	Counter Tuned, Sweep Track, AGen-Track (analog generator), DGen-Track (digital generator), or Fixed (set by direct entry)
Notch Tracking Range	±2.5% from fixed setting

#### **Crosstalk Function**

Frequency Range	10 Hz–200 kHz
Accuracy <sup>14</sup>	±0.4 dB, 20 Hz–120 kHz
Residual Crosstalk <sup>14</sup>	
10 Hz–20 kHz	≤ –140 or 1 μV
20 kHz–100 kHz	$\leq$ -126 dB or 2.5 $\mu$ V

#### **IMD Measurements**

with option "IMD"

Option "IMD" adds the capability to measure intermodulation distortion (IMD) using three of the most popular techniques. The demodulated IMD signal can also be selected for FFT analysis in SYS-2622 and SYS-2722 configurations.

#### SMPTE (DIN) IMD Function

Any combination of 40 Hz–250 Hz (LF) and 2 kHz–100 kHz (HF) tones, mixed in any ratio from 0:1 to 8:1 (LF:HF)
Amplitude modulation products of the HF tone. –3 dB measurement bandwidth is typically 20 Hz–750 Hz
0 to 20%
±0.5 dB
≤0.0015% [–96.5 dB], 60/7 kHz or 250/8 kHz

#### **CCIF and DFD IMD Functions**

Test Signal Compatibility	Any combination of equal amplitude tones from 4 kHz to 100 kHz spaced 80 Hz to 1 kHz
IMD Measured	
CCIF	2 <sup>nd</sup> order difference frequency product relative to the amplitude of either test tone
DFD	u₂ (2nd order difference frequency product) per IEC 268-3 (1986)

 $^{14}$  Uses the 1/3-octave bandpass filter to enhance the measured range in the presence of wideband noise. Alternate (interfering) channel input must be  $\geq 5$  mV.

 $^{15}$ System specification measured with the Cascade Plus analog generator at any valid input level  $\geq$ 200 mVrms.

Measurement Range	0 to 20%
Accuracy	±0.5 dB
Residual IMD <sup>15</sup>	CCIF ≤0.0004% [–108 dB], 14 kHz + 15 kHz, DFD ≤0.0002% [–114 dB], 14 kHz + 15 kHz

#### DIM (TIM) IMD Function

Test Signal Compatibility	2.96 kHz–3.15 kHz squarewave mixed with 14 kHz–15 kHz sine wave (probe tone)
IMD Measured <sup>16</sup>	$u_4$ and $u_5$ per IEC 268-3 (1986)
Measurement Range	0 to 20%
Accuracy	±0.7 dB
Residual IMD15	≤0.0020% [–94 dB]

#### **Wow & Flutter Measurements**

with option "W&F"

Option "W&F" adds the capability to make both conventional wow & flutter and scrape flutter measurements (using the technique developed by Dale Manquen of Altair Electronics, Inc.). The demodulated W&F signal can also be selected for FFT analysis in SYS-2622 and SYS-2722 configurations.

Test Signal Compatibility	
Normal	2.80 kHz–3.35 kHz
"High-band"	11.5 kHz–13.5 kHz
Measurement Range	0 to 1.2%
Accuracy (4 Hz)	±(5% of reading + 0.0005%)
Detection Modes	IEC/DIN (quasi-peak per IEC-386), NAB (average), JIS (per JIS 5551)
Response Selections	
Weighted	4 Hz bandpass per IEC/DIN/NAB
Unweighted	0.5 Hz–200 Hz
Scrape <sup>17</sup>	200 Hz–5 kHz
Wideband <sup>17</sup>	0.5 Hz–5 kHz
Residual W+F	
Weighted	≤0.001%
Unweighted	≤0.002%
Scrape or Wideband	≤0.005%
Minimum Input	5 mV, 20 mV for specified residual
Settling Time	
IEC/DIN or NAB	Typically 3 to 6 seconds
JIS	Typically 15 to 20 seconds

<sup>&</sup>lt;sup>16</sup> IEC 268-3 defines nine possible DIM products. The Cascade Plus IMD option analyzer is sensitive only to the u4 and u5 products using the simplified measurement technique proposed by Paul Skritek. DIM measurements using this technique will typically be 6–8 dB lower (better) than the results obtained using FFT-based techniques which sum all nine products.

<sup>&</sup>lt;sup>17</sup> Operational only with high-band test signals (11.5 kHz–13.5 kHz). Upper –3 dB rolloff is typically 4.5 kHz using 12.5 kHz.

# **Option Filters**

Up to seven option filters can be installed in the analog analyzer for weighted noise or other special measurements. Only one option filter may be enabled at a time, and it is cascaded with the standard bandwidth limiting filters. The following tables list only the most popular types. Consult Audio Precision for custom designs.

# **Weighted Noise Measurement**

FIL-AWT	"A" weighting per IEC Rec 179	see Figure 7
FIL-CCR	Weighting per IEC468 (CCIR) and (Also for CCIR/ARM)	DIN 45404 see Figure 8
FIL-CIT	Weighting per CCITT Rec P53	see Figure 9
FIL-CMS	"C-message" per BSTM 41004 an Std 743-1984	d ANSI/IEEE see Figure 10
FIL-CWT	"C" weighting per IEC Rec 179	see Figure 11

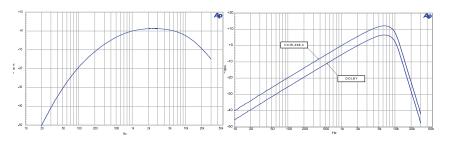


Figure 7. FIL-AWT. ANSI-IEC "A" Weighting Filter

Figure 8. FIL-CCR. IEC468 (CCIR)/ DIN 45404 Noise Weighting Filter

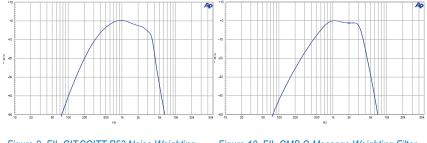
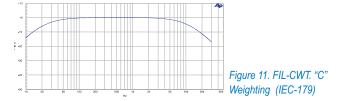


Figure 9. FIL-CIT.CCITT P53 Noise Weighting Filter





# **Precision De-emphasis Family**

FIL-D50	50 μs ±1%	see Figure 12
FIL-D50E	50 $\mu s$ $\pm 1\%$ + 15.625 kHz notch	
FIL-D50F	50 $\mu s$ $\pm 1\%$ + 19.0 kHz notch	see Figure 13
FIL-D75	75 μs ±1%	see Figure 14
FIL-D75B	75 $\mu$ s $\pm$ 1% + 15.734 kHz notch	see Figure 15
FIL-D75F	75 μs ±1% + 19.0 kHz notch	see Figure 16

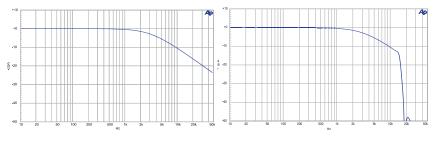


Figure 12. FIL-D50.50  $\mu s$  De-emphasis Filter.

Figure 13. FIL-D50F. 50  $\mu s$  with 19 kHz (FM) notch De-emphasis Filter.

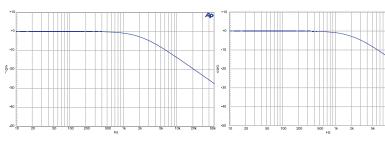


Figure 14. FIL-D75. 75  $\mu s$  De-emphasis Filter.

Figure 15. FIL-D75B. 75  $\mu s$  with 15.734 kHz (NTSC) notch De-emphasis Filter

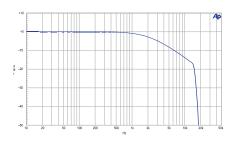
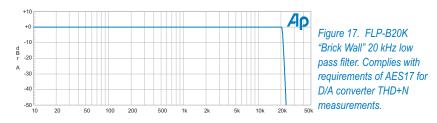


Figure 16. FIL-D75F. 75  $\mu s$  with 19 kHz (FM) notch De-emphasis Filter.

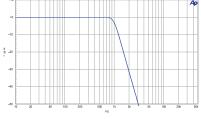
#### Very Sharp Cutoff Low-Pass Filter Family

FLP-B20K	±0.1 dB, 10 Hz–20 kHz; >60 dB attenuation at 24 kHz and higher. Complies with AES17. <i>see Figure 17</i>
FLP-B40K	±0.1 dB, 10 Hz–40 kHz; >60 dB attenuation at 48 kHz and higher. Complies with AES17.



# **General Purpose Low-Pass**

FLP-300	300 Hz ±3%, 5-pole	
FLP-400	400 Hz ±3%, 5-pole	
FLP-500	500 Hz ±3%, 5-pole	
FLP-1K	1 kHz ±3%, 5-pole	see Figure 18
FLP-3K	3 kHz ±3%, 7-pole	
FLP-4K	4 kHz ±3%, 7-pole	
FLP-8K	8 kHz ±3%, 7-pole	see Figure 19
FLP-50K	50 kHz ±5%, 3-pole	



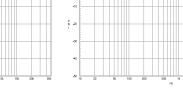
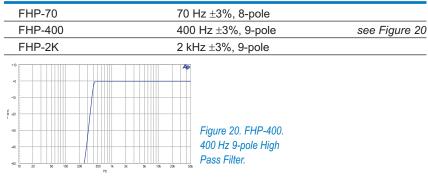


Figure 18. 1 kHz 5-pole Low Pass Filter.

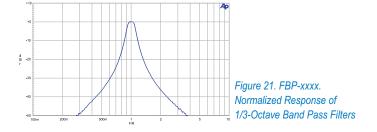


# **General Purpose High-Pass**



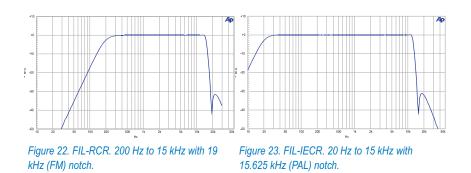
Family Response	Class II (4-pole) $\pm 0.2$ dB from 0.97 f <sub>o</sub> to 1.03 f <sub>o</sub> ;	
	<-12 dB at 0.8 f <sub>o</sub> and 1.25 f <sub>o</sub> ;	
	< -32 dB at 0.5 f <sub>o</sub> and 2.0 f <sub>o</sub>	see Figure 2
FBP-120	f <sub>o</sub> = 120 Hz	
FBP-250	f <sub>o</sub> = 250 Hz	
FBP-300	f <sub>o</sub> = 300 Hz	
FBP-400	f <sub>o</sub> = 400 Hz	
FBP-500	f <sub>o</sub> = 500 Hz	
FBP-600	f <sub>o</sub> = 600 Hz	
FBP-800	f <sub>o</sub> = 800 Hz	
FBP-1000	f <sub>o</sub> = 1.00 kHz	
FBP-1200	f <sub>o</sub> = 1.20 kHz	
FBP-1500	f <sub>o</sub> = 1.50 kHz	
FBP-2000	f <sub>o</sub> = 2.00 kHz	
FBP-3000	f <sub>o</sub> = 3.00 kHz	
FBP-4000	f <sub>o</sub> = 4.00 kHz	
FBP-5000	f <sub>o</sub> = 5.00 kHz	
FBP-6000	f <sub>o</sub> = 6.00 kHz	
FBP-8000	f <sub>o</sub> = 8.00 kHz	
FBP-10000	f <sub>o</sub> = 10.0 kHz	
FBP-12500	f <sub>o</sub> = 12.5 kHz	
FBP-15000	f <sub>o</sub> = 15.0 kHz	
FBP-20000	f <sub>o</sub> = 20.0 kHz	
FBP-30000	f <sub>o</sub> = 30.0 kHz	

# 1/3-Octave (Class II) Bandpass Family

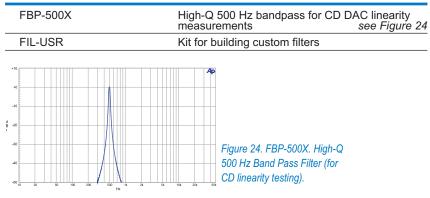


# **Receiver Testing**

FIL-RCR	200 Hz–15 kHz + 19.0 kHz notch	see Figure 22
FIL-IECR	20 Hz–15 kHz + 15.625 kHz notch	see Figure 23



#### **Miscellaneous**



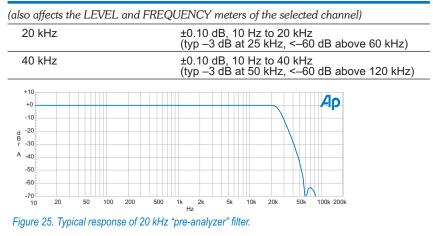
# **Option S-AES17**

Option S-AES17 adds the capability to insert a 20 kHz or 40 kHz low-pass filter following the selected analog input preamplifier, but before any signal processing within the analog analyzer. It enables accurate noise and THD+N measurements of sigma-delta converters and switching power amplifiers that contain large amounts of unwanted energy above the normal audio bandwidth.

High performance sigma-delta converters and switching power amplifiers often contain out-of-band energy that can exceed the in-band audio signal. Standard bandwidth limiting and noise weighting filters will not give accurate measurements due to their relatively low roll-off rates.

Option S-AES17 also includes the FLP-B20K and FLP-B40K option filters. These have been designed to work in tandem with the selectable pre-analyzer filters to provide THD+N measurements in accordance with AES17-1998. When the option is present, four new choices appear in the bandwidth (HF) drop down menu of the analog analyzer. "20k AES17" enables both the 20 kHz pre-analyzer filter and the FLP-B20K option filter. "20k SPCL" enables only the 20 kHz pre-analyzer filter for special applications such as weighted noise measurements using a different option filter. The "40k AES17" and "40k SPCL" choices provide similar functionality using the 40 kHz pre-analyzer filter and the FLP-B40K option filter. Enabling any of these four new choices will introduce a significant phase shift into the selected analog channel that will not be matched in the unselected channel. Thus phase measurements will be invalid whenever the pre-analyzer filter is active. Operation with the analog analyzer's standard "22 kHz," "30 kHz," "80 kHz," and ">500kHz" bandwidth limiting selections remains unchanged.

#### Pre-Analyzer Filter Response



#### **Residual THD+N (1 kHz)**

"20k AES17" mode	≤(0.00030% + 1.0 μV) [–110.5 dB]	
"40k AES17" mode	≤(0.00040% + 1.4 µV) [–108 dB]	

# **DSP** Analysis of Analog Signals

Available only in SYS-2622 and SYS-2722 configurations. Signals connected to the analog analyzer input connector may be routed through stereo A/D converters for enhanced analysis capabilities. There are two selectable converters. The high-resolution converter ("HiRes A/D") is optimized for signal analysis and FFT displays up to 30 kHz. It offers the best residual noise and distortion performance. The high bandwidth converter ("HiBW A/D") is optimized for signal analysis up to 120 kHz.

The term "SR" refers to sample rate, in hertz.

#### **High Resolution Converter**

A/D Resolution	24-bit sigma-delta
Sample Rate (SR)	7.2 ks/s to 108 ks/s variable; or 65.536 ks/s fixed
Flatness (1 kHz ref)	$\pm 0.01~\text{dB}$ to $0.45 \times SR$ or 20 kHz, whichever is lower
Alias Rejection <sup>18</sup>	typically >115 dB for signals >0.554 SR
Distortion	–105 dB for SR ≤65.536 ks/s, –102 dB for SR up to 100 ks/s

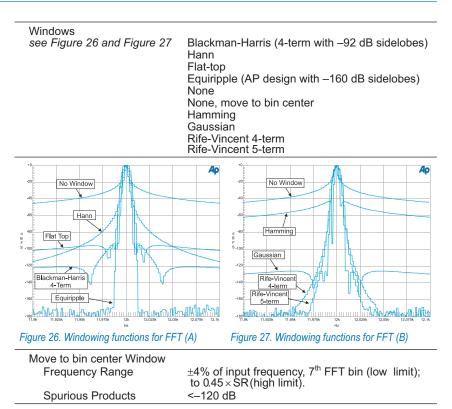
#### **High Bandwidth Converter**

16-bit sigma-delta
56 ks/s to 216 ks/s variable; or 131.072 ks/s or 262.144 ks/s fixed
±0.01 dB to 20 kHz, ±0.10 dB to 120 kHz (262.144 ks/s)
typically >85 dB for signals >0.540 × SR
–92 dB for SR ≤216 ks/s, –90 dB with SR = 262.144 ks/s

# **FFT Signal Analyzer**

(With "FFT" DSP program)	
Acquisition Length	800 to 4 M samples in 15 steps
Transform Length	256 to 32768 samples in binary steps
Processing	48 bit
Amplitude Accuracy	±0.05 dB, 20 Hz to 20 kHz Flat-top or Move to Bin Center windows
Averaging	1 to 4096 in binary steps. Averaging is power-based (frequency domain), or synchronous (time domain).
Waveform Display Modes	
Time Domain	Normal, Interpolate, Peak or Max
Frequency Domain	Peak pick (highest bin amplitude is displayed between the requested graph points)
Frequency Display Modes	Peak pick, individual bin

<sup>18</sup>Alias rejection is provided by digital filters within the A/D converters.



# **DSP Audio Analyzer**

with "Analyzer" DSP program

#### Wideband Level/Amplitude

Frequency Range	<10 Hz to 45% of sample rate [10 Hz to 21.6 kHz at 48 ks/s]
High pass Filters	<10 Hz 4-pole 22 Hz 4-pole 100 Hz 4-pole 400 Hz 4-pole (4-pole Butterworth or 10-pole elliptic if no other filters are enabled)
Low pass Filters	Fs/2 (maximum bandwidth) 20 kHz (6-pole elliptic) 15 kHz (6-pole elliptic)
Weighting Filters	ANSI-IEC "A" weighting, per IEC Rec 179 CCIR QPk per IEC468 (CCIR) CCIR RMS per AES17 C-message per IEEE Std 743-1978 CCITT per CCITT Rec. 0.41 "F" weighting corresponding to 15 phon loudness contour see Figure 28 HI-2 Harmonic weighting

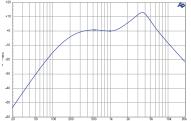
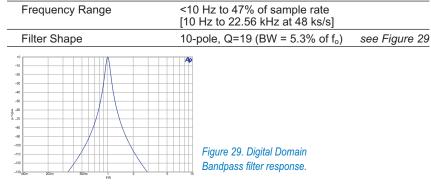


Figure 28. Digital Analyzer F-weighting curve.

#### Narrow Band Amplitude



#### THD+N Measurements

Frequency Range	<10 Hz to 47% of sample rate [10 Hz to 22.56 kHz at 48 ks/s]
High pass Filters	<10 Hz (4-pole) 22 Hz (4-pole) 100 Hz (4-pole) 400 Hz (4-pole Butterworth)
Low pass Filters	Fs/2 (maximum bandwidth) 20 kHz (6-pole elliptic) 15 kHz (6-pole elliptic)
Weighting Filters	ANSI-IEC "A" weighting, per IEC Rec 179 CCIR QPk per IEC468 (CCIR) CCIR RMS per AES17 C-message per IEEE Std 743-1978 CCITT per CCITT Rec. 0.41 "F" weighting corresponding to 15 phon loudness contour see Figure 29 HI-2 Harmonic weighting

#### Frequency Measurements

Range	<10 Hz to 47% of sample rate [10 Hz–23.0 kHz at 48 ks/s]
Accuracy	$\pm 0.01\%$ of reading or 0.0001% of sample rate, whichever is greater
Resolution	0.003% of reading or 0.0001% of sample rate, whichever is greater

#### Phase Measurements

Measurement Ranges	$\pm$ 180, –90/+270, or 0/+360 degrees
Accuracy <sup>19</sup>	
10 Hz–5 kHz	±0.5 degree
5 kHz–20 kHz	±1 degree
20 kHz–50 kHz	±2 degrees
Resolution	0.01 degree
Minimum Input	1 mV, both inputs

#### SMPTE IMD Measurements

Test Signal Compatibility	Any combination of 40 to 250 Hz (LF) and 2 kHz to 45% of sample rate (HF) tones, mixed in any ratio from 1:1 to 5:1 (LF:HF)
IMD Measured	Amplitude modulation products of the HF tone. –3dB measurement bandwidth is 10 Hz to 750 Hz.
Measurement Range	0 to 20%
Accuracy	±0.5 dB
Residual IMD <sup>20</sup>	≤0.0025%, 60 + 7 kHz or 250 + 8 kHz

# **Quasi-Anechoic Acoustical Tester**

Four pink sequences, four white sequences
(Sample rate ÷ 2000) to (sample rate ÷ 2)
1.465 Hz at 48.0 ks/s
32767 samples or 131071 samples
32768
half Hann Hann <240 Hz to >8 kHz <120 Hz to >16 kHz

<sup>&</sup>lt;sup>19</sup> Both analog analyzer input channels must have same coupling (ac or dc) selection, and both DSP analyzer input channels must have same coupling (ac or dc) selection. Accuracy is valid for any input signal amplitude ratio up to ±30 dB. Upper frequency range limited to 45% of sample rate.

<sup>&</sup>lt;sup>20</sup>System specification measured with the System Two Cascade Plus analog generator. Valid for input levels >200 mVrms.

Time Windows (percent of data record to transition from 0 to full	
amplitude)	<5%
	<10%
	<20%
	<30%
Averaging	1 to 4096 in binary steps. Averaging algorithm is synchronous.

# Multitone Audio Analyzer

With "FASTTEST" DSP program	
Acquisition Length	512 to 32768 samples in binary steps
Transform Length	512 to 32768 samples in binary steps
Processing	48 bit
Measurements	Level vs frequency (Response), Total distortion vs frequency, Noise vs frequency, Phase vs frequency, Crosstalk vs frequency, Masking curve
Frequency Resolution	(Sample Rate ÷ Transform Length) [1.465 Hz with SR = 48 ks/s & Transform Length = 32768]
Frequency Correction Range	±3%
Distortion	≤–115 dB

# **Digital Signal Generator**

Available only in the SYS-2700 and SYS-2722 configurations. The System Two Cascade Plus digital generator consists of a DSP signal generator, selectable pre-emphasis filters, two hardware dither generators, and several digital output stages supporting the most popular formats.

Except for arbitrary waveforms, the digital outputs and the digitally generated analog signals are independently selectable and concurrently available. If both digital and analog outputs are selecting arbitrary waveform, it must be the same one.

#### **Output Formats** AES/EBU (per AES3-1992) SPDIF-EIAJ per IEC 60958 Optical (Toslink®) per IEC 60958 General purpose serial General purpose parallel Serial interface to chip level via optional SIA-2722 27 kHz-108 kHz AES/EBU, 54 kHz-216 kHz dual Sample Rates connector AES/EBU, general purpose serial; 8 kHz to 216 kHz parallel; independent of input sample rate Sample Rate Resolution 1/64 Hz (approx. 0.0156 Hz) Sample Rate Accuracy ±0.0002% [±2 PPM] using internal reference, lockable to external reference Word Width 8 to 24 bits Encoding Linear, µ-Law, A-Law Nominal Output Impedance Balanced (XLR) **110** Ω Unbalanced (BNC) 75 Q

# **Digital Output Characteristics**

# **Digital Signal Ceneration**

#### Sine Family Common Characteristics

Waveforms	Sine, Sine Burst (rectangular envelope), Variable Phase Sine (two sine waves of same frequency but settable phase), Stereo Sine (independent frequency and amplitude in each channel), Dual Sine (sum of two sine waves with variable ratio), Sine + Offset, and Shaped Sine Burst (raised cosine envelope)
Frequency Range	10 Hz to <50% of Sample Rate [<24 kHz at 48 ks/s]
Frequency Resolution	Sample Rate ÷ 2 <sup>23</sup> [0.006 Hz at 48 ks/s]
Flatness	±0.001 dB
Harmonics/Spurious Products	≤0.000001% [–160 dB]

#### Variable Phase Sine Wave

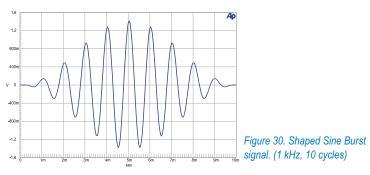
Phase Range	±180 deg.	
Phase Resolution	0.01 deg.	

#### Sine + Offset

Oliset Allplitude = Oliset allplitude + Oliset allplitude = 2100 /01 s	Offset Amplitude	Sine amplitude +  offset amplitude  ≤100% Fs
------------------------------------------------------------------------	------------------	----------------------------------------------

#### Sine Burst and Shaped Sine Burst

Envelope	Rectangular for Sine Burst,	
	Raised cosine for Shaped Burst	see Figure 30
Interval	2 to 65536 cycles	
Burst On	1 to (number of Interval cycles mir	านร 1)



#### Square Wave

Frequency Range	$\leq$ 1 Hz to 1/6 sample rate. Frequencies are limited to even integer sub-multiples of the Sample Rate.
Even Harmonic Content	≤0.000001% [–160 dB]

#### SMPTE/DIN Waveform

le rate
at 4:1 ratio

# CCIF and DFD IMD Waveforms

Center Frequency Range	3000 Hz to (<50% of sample rate $-\frac{1}{2}$ IM freq.)
IM Frequency Range	80 Hz–2.00 kHz
Distortion/Spurious	≤0.000001% [–160 dB]

#### DIM IMD Waveform

Square/Sine Frequencies	Determined by Sample Rate (see Note below)
Distortion/Spurious	≤0.000001% [–160 dB]

The DIM test signal consists of a square wave and a sine wave mixed in a 4:1 amplitude ratio. Since digital square waves are generated by alternately turning the output on and off for the same number of sample periods, the frequencies achievable are limited to even sub-multiples of the Sample Rate. Because of this constraint, the square wave frequency is chosen first to be as close to the "ideal" analog test frequency as possible. The sine wave frequency is then chosen based upon the ideal sine/square frequency ratio. The following table lists some examples for the DIM and DIMB signals:

DIM: "ideal" square frequency = 3150, sine/square frequency ratio = 100/21		
Sample Rate	Square Wave Frequency	Sine Wave Frequency
44100	3150	15000
48000	3000	14285.7
DIMB: "ideal' square frequency = 2960, sine/square frequency ratio = 175/37		
44100	3150	14898.65
48000	3000	14189.19

#### Noise

	Types	Pink, white, burst, USASI
--	-------	---------------------------

#### Special Signals

Low level staircase waveform for D/A linearity testing
Produces a maximum amount of data-induced jitter on low-bandwidth transmission links
Two sinewaves phased for reinforcement with normal polarity
A single binary one value "walked" from LSB to MSB
A single binary zero value "walked" from LSB to MSB
(Digital DC)
Random binary states of all bits
Passes the signal from the rear panel Ref Input. Accepts sample rates from 27 kHz to 100 kHz and outputs at programmed sample rate. Ratio of rates may not exceed 8:1.
32 bit when using triangular dither

#### Quasi-Anechoic Acoustical Tester (MLS)

(Also see MLS in Digital Analyzer section, page 33)

Signals	Four pink sequences, four white sequences
Frequency Range	DC to 50% of sample rate
Sequence Length	32767 samples or 131071 samples, automatically selected between 32 k or 131 k sequence as supplied by generator

#### Multitone Signals

Stored waveform consisting of multiple sine waves, each at independent frequency, amplitude, and phase

Maximum Number of Tones	Up to 8191 (maximum length)
Frequency Range	DC to Sample Rate ÷ 2
Frequency Resolution	Sample Rate ÷ 2 <sup>14</sup> (typically 2.93 Hz at 48 ks/s)

#### Arbitrary Waveforms ("Arb Wfm")

Signal	Determined by the associated file specified in the panel drop-down box.
Length	256 to 16384 points per channel. Utility is provided to prepare waveform from user specified frequency, amplitude, and phase data.
Frequency Resolution	Sample Rate ÷ Length [typically 2.93 Hz at 48 ks/s]

#### Dither

(may be enabled for all waveforms except Monotonicity, J-Test, Walking Ones and Zeroes, and Random)

Probability Distribution	Triangular or rectangular; true random; independent for each channel
Spectral Distribution	Flat (white) or Shaped (+6 dB/oct)
Amplitude	8 to 24 bit, or off

#### **Pre-Emphasis Filters**

(all waveforms)	
Filter Shape	50/15 μs or J17
Response Accuracy	$\pm 0.02$ dB, 10 Hz to 45% of Sample Rate
Residual Distortion	≤0.00003% [–130 dB]

# **AES/EBU Interface Generation**

# **Interface Signal**

Amplitude Range	
Balanced (XLR)	0 to 10.16 Vpp, $\pm$ (10% + 80 mV) into 110 $\Omega$ . 4 mV steps below 1 Vpp, 40 mV steps above.
Unbalanced (BNC)	0 to 2.54 Vpp, $\pm$ (8% + 20 mV) into 75 $\Omega$ . 1 mV steps below 0.25 Vpp, 10 mV steps above.
Optical (Toslink <sup>®</sup> )	0 to 256% of nominal intensity in 1% steps
Channel Status Bits	Full implementation per IEC 60958, English language decoded, Professional or consumer or hex formats; independent in each channel
User Bits	set to 0
Validity Flag	selectable, set or cleared

# **AES/EBU Impairments**

Variable rise/fall time	16 ns–400 ns, ±20%
Induced Jitter	Selectable sinewave, squarewave, or wideband noise
Jitter Freq Range <sup>21</sup>	2.00 Hz–200 kHz, <0.1 Hz resolution
Jitter Ampl Range <sup>21</sup>	0–1.27 UI (peak) in 0.005 UI steps; 1.3–12.7 UI (peak) in 0.05 UI steps
Jitter Accuracy	±(10% + 0.005 UI)
Jitter Flatness <sup>22</sup>	±1 dB, 100 Hz–20 kHz
Residual Jitter <sup>23</sup> 48 ks/s 96 ks/s	≤0.010 UI [1.6 ns] ≤0.020 UI [1.6 ns]
Spurious Jitter Products	typically 30 dB below jitter signal or <0.001 UI, whichever is larger
Normal Mode Noise Balanced Unbalanced	0 to 2.55 Vpp, in 10 mV steps; ±(10% + 100 mV) 0 to 635 mVpp, in 2.5 mV steps; ±(10% + 25 mV)
Common Mode Freq	20 Hz–40 kHz, <0.1 Hz resolution
Common Mode Ampl	0 to 20 Vpp, in 80 mV steps, ±(10% + 200 mV)
Cable Simulation	Multi-pole fit to AES3-1992 filter to simulate the response degradation of a long cable.
Offset from reference	-64 to +63.5 UI, in 0.5 UI steps

<sup>&</sup>lt;sup>21</sup> Combinations of jitter amplitude and frequency must not result in greater than 50% reduction in transmitted bit width.

<sup>&</sup>lt;sup>22</sup> System specification including generator and analyzer contributions valid only at 32.0, 44.1, 48.0, 64.0, 88.2, and 96.0 ks/s only. Flatness may be degraded at other sample rates

<sup>&</sup>lt;sup>23</sup> System specification including analyzer contribution. The following conditions must be met: (1) the jitter generator amplitude must be turned off or set for 0.0000 UI, (2) all other forms of impairment must be off or disabled, and (3) the digital output must be ≥1.0 Vpp (XLR) or ≥250 mVpp (BNC).

# **Reference Input Characteristics**

A rear panel reference input is provided to synchronize the internal sample clock generator to an external signal. The internal sample rate (ISR) is not dependent upon the rate or characteristics of the external reference. OSR need not be at 1:1 ratio to reference but will be phase-locked to reference over full specified range of OSR and Reference inputs. Phase lock loop bandwidth is approximately 5 Hz.

Input Formats	AES/EBU (per AES3-1992), NTSC/PAL/SECAM video, or squarewave
Input Sample Rates/ Frequency Range	28.8 kHz–100 kHz AES/EBU, 8.0 kHz–10.0 MHz squarewave
Sample Rate Resolution 8 kHz–65 kHz 65 kHz–256 kHz 256 kHz–1 MHz 1 MHz–4 MHz 4 MHz–10 MHz	1/128 Hz [0.0078125 Hz] 1/32 Hz [0.03125 Hz] 1/8 Hz [0.125 Hz] 1/2 Hz [0.5 Hz] 2 Hz
Minimum Input Amplitude	200 mVpp
Nominal Input Impedance AES/EBU (XLR) Video, square wave (BNC)	110 Ω or >5 kΩ 75 Ω or >5 kΩ
Lock Range	±0.0015% [±15 PPM]
Input Delay from Reference Display	Measures delay from 0 to 127.9 UI in seconds, $\pm 60$ ns
Reference Rate Display	Measures approximate reference input rate

# **Reference Output Characteristics**

A rear panel reference output is provided to drive devices under test that require their own reference input. The reference output signal is not jittered.

Output Format	AES/EBU (per AES11-1994)
Output Sample Rates	28.8 kHz–100 kHz AES/EBU; locked to front panel output
Status Bits	Format "Professional" Sample Rate indicates closest rate Type "Grade 2 reference" Origin "SYS2" Reliability flags implemented CRCC implemented Time of Day not implemented Sample Count not implemented
Output Delay from Reference Output	–64/+63.5 UI, in 0.5 UI steps; ±(5% + 0.5 UI)
Residual jitter	≤0.005 UI pk (120 Hz–100 kHz BW)

# **Digital Analyzer**

Available only in the SYS-2700 and SYS-2722 configurations.

# **Digital Input Characteristics**

Input Formats	AES/EBU (per AES3-1992) Dual Connector AES/EBU SPDIF-EIAJ per IEC-60958 Dual Connector SPDIF-EIAJ Optical (Toslink®) per IEC-60958 General purpose serial General purpose parallel Serial interface to chip level via optional SIA-2722
Sample Rates	28.8 kHz–100 kHz AES/EBU, 64 kHz–200 kHz Dual Connector AES/EBU, 8 kHz to 200 kHz parallel or via SIA-2722 (independent of output sample rate)
Word Width	8 to 24 bits
Nominal Input impedance	
AES/EBU	110 Ω or ≥2.5 kΩ
SPDIF-EIAJ	75 Ω or ≥3 kΩ

# **Embedded Audio Measurements**

With "Analyzer" DSP program

#### Wideband Level/Amplitude

Range	–120 dBFS to 0 dBFS (usable to –140 dBFS)
Frequency Range	<10 Hz to 45% of sample rate [10 Hz–21.6 kHz at 48 ks/s]
Accuracy	±0.01 dB
Flatness	±0.01 dB, 15 Hz–22 kHz (<10 Hz high-pass filter selection)
High pass Filters	<10 Hz (4-pole) 22 Hz (4-pole) 100 Hz (4-pole) 400 Hz (4-pole Butterworth, or 10-pole elliptic if no other filters are enabled)
Low pass Filters	Fs/2 (maximum bandwidth) 20 kHz (6-pole elliptic) 15 kHz (6-pole elliptic)
Weighting Filters	ANSI-IEC "A" weighting, per IEC Rec 179 CCIR QPk per IEC468 (CCIR) CCIR RMS per AES17 C-message per IEEE Std 743-1978 CCITT per CCITT Rec. 0.41 "F" weighting corresponding to 15 phon loudness contour see Figure 28, page 21 HI-2 Harmonic weighting
Residual Noise (at 48 ks/s and 96 ks/s SR)	–141 dBFS unweighted –144 dBFS A-weighted

–140 dBFS CCI	R RMS
-130 dBFS CCI	R QPk
–142 dBFS 20 k	Hz LP
–143 dBFS 15 k	Hz LP
–139 dBFS "F" v	veighting
-152 dBFS CCI	
–151 dBFS C M	essage

#### Narrow Band Amplitude

Frequency Range	<10 Hz to 47% of sample rate [10 Hz to 22.56 kHz at 48 ks/s]
Filter Shape	10-pole, Q=19 (BW = 5.3% of f <sub>o</sub> ) see Figure 29, page 21
Residual Distortion	≤–150 dBFS

#### THD+N Measurements

Frequency Range		<10 Hz to 47% of sample rate [10 Hz to 22.56 kHz at 48 ks/s]	
Residual THD+N	≤–138 dBFS	see Figure 31, page 35	
High pass Filters	<10 Hz (4-pole) 22 Hz (4-pole) 100 Hz (4-pole) 400 Hz (4-pole Butterwo	orth)	
Low pass Filters	Fs/2 (maximum bandwi 20 kHz (6-pole elliptic) 15 kHz (6-pole elliptic)		
Weighting Filters	Same as Wideband Lev	Same as Wideband Level/Amplitude	
Residual Noise	Same as Wideband Lev	vel/Amplitude	

#### **Frequency Measurements**

Range	<10 Hz to 47% of sample rate [<10 Hz–22.56 kHz at 48.0 ks/s]
Accuracy	$\pm 0.01\%$ of reading or 0.0001% of sample rate, whichever is greater
Resolution	0.003% of reading or 0.0001% of sample rate, whichever is greater

#### **Phase Measurements**

Measurement Ranges	$\pm$ 180, –90/+270, or 0/+360 degrees
Accuracy <sup>24</sup>	$\pm 2$ degrees, 10 Hz to 45% of Sample Rate
Resolution	0.01 degree
Minimum Input	–60 dBFS, both inputs

<sup>&</sup>lt;sup>24</sup>Both DSP analyzer input channels must have the same coupling (ac or dc) selection.

#### SMPTE IMD Measurements

Test Signal Compatibility	Any combination of 40 to 250 Hz (LF) and (2 kHz to <50% of sample rate) (HF) tones, mixed in any ratio from 1:1 to 5:1 (LF:HF)
IMD Measured	Amplitude modulation products of the HF tone. (–3 dB measurement bandwidth is typically 20 Hz–750 Hz.)
Measurement Range	0 to 20%
Accuracy	±0.5 dB
Residual IMD	≤–130 dB at 0 dBFS, 60 + 7 kHz or 250 + 8 kHz ≤–110 dB at –25 dBFS, 60 + 7 kHz or 250 + 8 kHz

#### FFT Spectrum Analyzer

with "FFT" DSP program (48 bit j	processing)
Acquisition Length	800 to 256 k samples in 11 steps
Transform Length	256 to 32768 samples in binary steps
Processing	48 bit
Windows (see Figures 26 and 27, page 20)	Blackman-Harris (4-term with –92 dB sidelobes) Hann Flat-top Equiripple (–160 dB sidelobes) None None, move to bin center Hamming Gaussian Rife-Vincent 4-term Rife-Vincent 5-term
Amplitude Accuracy	±0.001 dB, 20 Hz to 20 kHz, with Flat-top window
Phase Accuracy <sup>25</sup>	$\pm 0.05$ degree, 10 Hz to 45% of Sample Rate
Resolution	0.01 degree
Averaging	1 to 4096 in binary steps. Averaging is power-based (frequency domain), or synchronous (time domain)
Distortion Products	≤–160 dB
Frequency Display Modes Time Domain Frequency Domain	Normal, interpolate, peak or max Peak pick, individual bin
Move to bin center Window Frequency Range Amplitude Accuracy Spurious Products	±4% of input frequency, 7 <sup>th</sup> FFT bin (low limit); to 0.45 × SR (high limit). ±0.025 dB ≤–120 dB

 $<sup>^{25}</sup>$  Both dsp analyzer input channels must have same coupling (ac or dc) selection. Accuracy is valid for any input signal amplitude ratio up to  $\pm 30$  dB.

# Multi-Tone Audio Analyzer

with "FASTTEST" DSP program (48 bit processing)		
Acquisition Length	512 to 32768 samples in binary steps	
Transform Length	512 to 32768 samples in binary steps	
Processing	48 bit	
Measurements	Level vs frequency, Total distortion vs frequency, Noise vs frequency, Phase vs frequency, Crosstalk vs frequency, Masking curve	
Frequency Resolution	Sample Rate ÷ 2 <sup>15</sup> [1.465 Hz with 48.0 ks/s]	
Frequency Correction Range	±3%	
Distortion	≤–140 dB	

with "FASTTEST" DSP program (48 bit processing)

# **Quasi-Anechoic Acoustical Tester**

with "MLS" DSP program	
Signals	Four pink sequences, four white sequences
Frequency Range	Sample rate/2000 to sample rate/2
Frequency Resolution (Max)	1.465 Hz at 48.0 ks/s
Acquisition Length	32767 samples, 131071 samples, automatically selected between 32 k or 131 k sequence as supplied by generator
FFT Length	32768
Energy Time Windows	half Hann Hann <240 Hz to >8 kHz <120 Hz to >16 kHz
Time Windows (percent of data record to transition from 0 to full amplitude)	<5% <10% <20% <30%
Averaging	1 to 4096 in binary steps, synchronous

# **Digital Interface Analyzer**

#### with "INTERVU" DSP program

INTERVU operates in conjunction with an autoranged 8-bit A/D converter clocked at 80.0 MHz, providing interface signal measurements with >30 MHz bandwidth. INTERVU can display the interface signal in time or frequency domain, as an eye pattern, or as probability graphs of amplitude or pulse width. INTERVU also can demodulate the jitter signal and display it in the time or frequency domain or as a histogram. The jitter signal or the data on the interface may be reproduced through the monitor loudspeaker.

AES/EBU Input Voltage	
Balanced	0 to 20.48 Vpp, ±(10% + 50 mV)
Unbalanced	0 to 4.096 Vpp, ±(8% + 12 mV)
Jitter Amplitude	0 to 5 UI pk, ±(5% + 0.015 UI)
Residual Jitter	≤0.01 UI (50 Hz–1 MHz BW)
Spurious Jitter Products	≤0.001 UI, or ≤–60 dB below jitter signal
Common Mode Amplitude	0 to 20.48 Vpp, ±(30% + 50 mV), 20 kHz–1 MHz
Jitter Probability Display	256 bins, autoranging
Input Probability Display	256 bins, autoranging
Bit Width Probability Display	32768 bins
Risetime	≤20 ns
Acquisition time / memory	19.66 ms / 1,572,864 samples

# **Digital Interface Measurements**

#### AES/EBU Impairments, real time displays

Input Sample Rate	±0.0003% [±3 ppm] internal reference, ±0.0001% [±1 ppm] external reference
Output to Input Delay	Measures status propagation from the AES/EBU output to the input. Range is 0 to 1 frame, resolution ±60 ns.
AES/EBU Input Voltage	
XLR	100 mV to 10.16 Vpp, ±(5% + 50 mV)
BNC	50 mV to 2.54 Vpp, ±(5% + 12 mV)
Jitter Amplitude <sup>26</sup>	
50 Hz–100 kHz BW	0 to 3.00 UI, ±(10% + 0.01 UI)
Other BW selections	0 to 1.00 UI, ±(10% + 0.005 UI)
Jitter Flatness <sup>27</sup>	±1 dB, 100 Hz–20 kHz
Residual Jitter <sup>28</sup>	≤1.6 ns [0.010 UI at 48 ks/s, 0.020 UI at 96 ks/s]

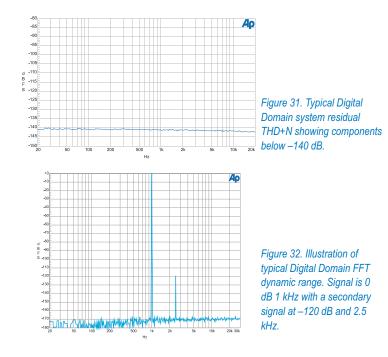
<sup>&</sup>lt;sup>26</sup> Jitter amplitude is peak calibrated.

<sup>&</sup>lt;sup>27</sup> System specification including generator and analyzer contributions at 32.0, 44.1, 48.0, 64.0, 88.2, and 96.0 ks/s only. Flatness may be degraded at other sample rates.

<sup>&</sup>lt;sup>28</sup> System specification including generator contribution. The following conditions must be met: (1) the jitter generator amplitude must be turned off or set for 0.0000 UI, (2) all other forms of impairment must be off or disabled, and (3) the digital input must be ≥1.0 Vpp (XLR) or ≥250 mVpp (BNC).

Jitter Spectrum	Spurious products are typically 40 dB below jitter signal or <0.0003 UI [–70 dBUI], whichever is larger
Common Mode Ampl	0 to 20.48 Vpp, ±(10% + 300 mV), 315 Hz–200 kHz
Cable Equalization	Per AES3-1992
Channel Status Bits	Full implementation, English language decoded (Professional or Consumer) hex formats, independent in each channel
User Bits	Not displayed
Validity Flag	Displayed for each channel
Parity	Displayed for total signal (both channels combined)
Signal Confidence	Displayed for total signal (both channels combined)
Receiver Lock	Displayed for total signal (both channels combined)
Coding Error	Displayed for total signal (both channels combined)

# **Graphs of Typical Digital Domain Performance**



# **Auxiliary Signals**



Figure 33. Monitors panel

# **Generator Signal Monitors**

(All units except SYS-2700. See Figure 33)	
Channel A	Buffered version of the channel A analog generator signal. Amplitude is typically 2.8 Vpp.
Channel B	Buffered version of the channel B analog generator signal. Amplitude is typically 2.8 Vpp.

# **Generator Auxiliary Signals**

(All units except SYS-2700. See Figure 33)	
Sync Output	LSTTL compatible signal that is intended to be used as a trigger for stable oscilloscope displays.
Trig/Gate Input	LSTTL compatible input, functional with option "BUR" only.

# **Analyzer Signal Monitors**

(All units except SYS-2700. See Figure 33)

Channel A	Buffered version of the channel A analog input signal. Amplitude is typically 0 to 3.6 Vpp.
Channel B	Buffered version of the channel B analog input signal. Amplitude is typically 0 to 3.6 Vpp.
Reading	Buffered version of the analog analyzer output signal after all filtering and gain stages. Amplitude is typically 0 to 3.6 Vpp.

# **Digital Signal Monitors**

(SYS-2700.& SYS-2722 only. See Figure 33)

Via four 24-bit D/A converters. Function monitored depends upon analyzer program loaded; for example, noise and distortion products after notch filter are monitored with "DSP Audio Analyzer" in its THD+N function.

Channel 1	Buffered version of the digital channel 1 signal
Channel 2	Buffered version of the digital channel 2 signal
Reading 1	Distortion of the digital channel 1 signal
Reading 2	Distortion of the digital channel 2 signal



**Digital Interface Monitors** 

(SYS-2700.& SYS-2722 only. See Figure 34)

Transmit Frame Sync	Squarewave at the programmed internal sample rate
Receive Frame Sync	Squarewave at the rate of the received AES/EBU signal
Master Clock Out	A squarewave at a multiple of the programmed output sample rate (SRO). The multiple is 1024x for sample rates of 6.8 kHz–12 kHz; 512x for sample rates of 12 kHz–24 kHz; and 256x for sample rates of 24 kHz–96 kHz. Selectable between jittered and unjittered signals.

# Miscellaneous Digital I/O

(SYS-2700.& SYS-2722 only. See Figure 34)	
Auxiliary Input	LSTTL compatible trigger input for DSP program data acquisition
Auxiliary Output	HCMOS signal, function under DSP program control
Trigger Output	HCMOS signal, coincident with period of generated signal waveform

# **Audio Monitor**

All configurations contain an internal loudspeaker and headphone jack for listening to the generator, analyzer, or digital signal monitor points, including noise and distortion following analog or digital notch filters or the AES/EBU jitter signal. Use of the audio monitor does not preclude the use of any measurements.

|--|

# General/Environmental

Power Requirements	100/120/230/240 Vac (-10%/+6%), 50/60 Hz, 240 VA max
Temperature Range	
Operating	+5°C to +40°C
Storage	–40°C to +75°C
Humidity	90% RH to at least +40°C (non-condensing)
Altitude	2000 m (operating)
EMC <sup>29</sup>	Complies with 89/336/EEC, EN 61326-1 Class B/CISPR 22, and FCC 15 subpart J (class B)
Dimensions	
Width	41.9 cm [16.5 inches]
Height	14.6 cm [5.75 inches] bench-top (with feet) 3U [5.25 inches] rack-mount
Depth	34.5 cm [13.6 inches]
Weight	Approximately 15.4 kg [34 lbs]
Safety	Complies with 73/23/EEC and 93/68/EEC. EN61010-1 (1990) + Amendment 1 (1992) + Amendment 2 (1995) Installation Category II—Pollution Degree 2.

<sup>&</sup>lt;sup>29</sup>Emission and immunity levels are influenced by the shielding performance of the connecting cables. The shielding performance of the cables will depend on the internal design of the cable, connector quality, and the assembly methods used. EMC compliance was demonstrated using Audio Precision cables CAB-XMF and CAB-AES2.



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