

Audio III precision

Fundamentals of audio test





⇒ audio test fundamentals

- ⇒ measuring modern audio
- how to avoid the most common mistakes in audio test



Who is Audio Precision

The recognized standard in audio test

Since 1985 AP has provided audio test and measurement instruments to R&D and production lines for every type of audio device.

AP is the number one maker of audio analyzers in the world.







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Who uses AP?



- → Pro audio
- A/V receivers
- ⇒ Smartphones
- ⇒ Semiconductors
- ⇒ Loudspeakers
- ➡ Microphones
- ⇒ Automotive head units
- ⇒ Blu-ray players
- → MP3 players
- ⇒ Bluetooth headsets
- ⇒ Tablets and PCs
- → Hearing aids
- → Telecom
- ➡ Military





mmm



Why: Improve quality

☑ Good test allows higher quality, which sells better than lower quality





Why: Reduce cost

Reduce costs from re-designs, recalls and returns by catching mistakes before they become expensive





Why: Save time

Verify new designs for speed to market and make sure faults are found quickly to keep production running smoothly







Why: Trust & reliability

Evaluate components you're considering for your product and test work done by subcontractors





Audio is different

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Extremely wide dynamic range of signals

- Electronic analog of our range of hearing can range from 0 dBspl to >130 dBspl
- ⇒ 80 dB to over 100 dB range is common
- ⇒ Levels from microVolts to hundreds of Volts

☑ Wide frequency range

- ⇒ 20 Hz 20 kHz common (1,000:1, 10 octaves),
- ⇒ 10 Hz- 100 kHz (10,000:1, 13+ octaves)
- ➡ If this was radio this would be like 1 MHz to 1 GHz from the AM radio band to microwave!

For every +6dB, double the voltage... For every -6dB, half the voltage



It's always analog somewhere

- ☑ Analog audio
- ☑ Digital audio
- ☑ Cross domain analog and digital
- ✓ "Third domain" digital interface and metadata





Common Measurements: The Big 6

	dBV	Volts
1 Lovel & Gain	+20	10.0
	+18	7.94
⇒ Level: Amplitude of an audio signal, expressed in dB.	+12	3.98
⇒ Gain: Relative level of output vs. input	+6	1.99
	0 dBV	1 Volt
	-6	0.50
	-12	0.25
Output level (Volts)	-18	0.125
= Gain (dB)	-20	0.1
	-40	0.01
	-60	0.001
	-80	0.0001
	-100	0.00001
Scope (Signal)		125
Signal Acadetion and Analyss © Show Signal © Show THD-N Residual	ater the ampl al, the higher	itude of the level
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200u 400u 600u 800u 1.0m 1.2m 1.4m 1.6m 1.3m 2.0m 2.2m 2.4m 2.5m 2.8m Time (s) Interpolation: ⊙ Ch ◯ Off		> ap.cor



Common Measurements: The Big 6

② THD+N: Total Harmonic Distortion plus noise

- ⇒ Summation of all distortion and noise ÷ Total level
- ⇒ Usually a single frequency measurement
- ⇒ .0001 Vrms (noise+distortion) ÷ 1 Vrms (total level) = 0.01% or -80 dB at 1 kHz



Fundamental

Harmonics

Undesired extra energy at multiples of the fundamental

1 kHz =>2 kHz... 3 kHz... 4 kHz... etc 4 kHz =>8 kHz... 12 kHz... 16 kHz... etc



Common Measurements: The Big 6

Frequency Response - "flatness"

- ⇒ Amplitude versus Frequency, usually a specification
- \Rightarrow "Response ±2 dB 20-20kHz", sometimes a graph

☑ Signal-to-Noise Ratio

- ⇒ in dB below a maximum or reference level
- SNR = 96 dB A weighted

Phase

- ⇒ Phase between the two channels a graph or a spec. across a frequency band
- ⇒ "± 2 Deg. 20 20kHz", sometimes a single number at high frequency
 (10kHz)
- ⇒ Stereo "image" is directly affected by phase error

Crosstalk

- A measurement at a single frequency or a graph of Amplitude versus Frequency.
- ⇔ "Crosstalk <85 dB"</p>



- Maximum output
- **Equalized frequency response**
- ☑ Weighted Noise, Equivalent Input Noise
- ☑ Individual Harmonic amplitude odd, even, 3rd etc.
- ☑ Inter Modulation Distortion SMPTE, CCIF, DIM/TIM
- ☑ Input-Output Phase, Group Delay, Polarity inversion
- ☑ Gain Linearity
- ☑ Gain or Loss, Input & Output impedance
- ☑ Noise based measurements pink, white, band-limited





Digital Audio

✓ ALL the analog measurement types but within the digital data stream:

- ⇒ Big Six and variations on these
- ☑ The interface or transport itself
- ✓ More likely to be a problem with metadata

Bit	If not set means:	If set means:	
0	Consumer (S/PDIF)	Professional (AES3) (changes meaning of control word)	
1	Normal Compressed data		
2	Copy restrict Copy permit		
3	2 channels	channels 4 channels	
4	– –		
5	No pre-emphasis Pre-emphasis		
6–7	Mode, defines subsequent bytes, always zero		
8–14	Audio source category (general, CD-DA, DVD, etc.)		
15	L-bit, original or copy (see text)		

File I Hex I CEA Audio Data Block I Short Audio Descriptor 1 Linear PCM Audio Format Linear PCM Number of Channels (1-8) Sampling Rate: 192 kHz Sampling Rate: 176.4 kHz Sampling Rate: 96 kHz Sampling Rate: 88.2 kHz Sampling Rate: 48 kHz Sampling Rate: 32 kHz Sampling Rate: 32 kHz Sampling Rate: 32 kHz Sampling Rate: 32 kHz Bit Depth: 20 bit Bit Depth: 16 bit	-EDID Viewer		
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Bit Depth: 16 bit True	Bit Depth: 20 bit	True	
DR DODAL TO DR	Bit Depth: 16 bit	True	



Digital metadata

	Metadata
	AD
Lock Status	Locked
Measured Sample Rate	e 48 95 191940 kHz 44.09630 kHz
IEC61937 data-type	PCM
S8:Application(A)	
SB:Audio Mode(A)	Audio
SB:Category Code(A)	
StriSampling Frequency(A)	44.1 kHz 48 kHz 96 kHz 96 kHz 192 44.1 kHz 48 96 kHz 192 kHz 44.1 kHz 44.1 kHz
S8:Word Length(A)	20 Bits 24 Bits 20 Bits
HD:A/V Mute	False
HD:HDCP Decrypting	False
HD:Channel Count	t 2ch 6ch XX 6ch
HD:Sample Size	Refer to Stream Header
HD:SpeakerAllocation	
HD:Coding Type	Refer to Stream Header
HDitevel Shift Value	0 dB
HD:HDCP Authentication	
	1.800 1.905 8.000 8.200 13.200 13.400 19.400 19.500 24.100 24.300 57.500 57.600 62.700 62.900 79.795 98.641 98.800 99.000 109.972 112.288 120.0 Transition Time(s)





▲ s(t)

Some basic digital concepts

✓ Sample rate ⇒ How many times a second a digital sample is taken ✓ Nyquist frequency:

➡ ½ the sample rate equals maximum frequency that can be captured

Common Sample rates8 kHzTelecom16 kHz"Wideband"voice44.1 kHzCompact disc48 kHzPro audio96 kHzStereophile

192 kHz HDMI





Some basic digital concepts





One more concept: FFTs

☑ Fast Fourier Transform

- ⇒ The basis of modern audio analyzers
- ⇒ View signals in frequency domain
- ⇒ Easy way to see distortion and noise
- ⇒ Can derive most other measurements mathematically





One more concept: FFTs

☑ Not all FFTs are equal

⇒ Bin width and bit depth



Common digital audio interfaces



☑ Consumer

- ⇒ PCM, HDMI, Bluetooth,
- ⇒ On the way out: SPDIF
- Pro audio
 - AES/EBU, MADI
 - ⇒ On the way up: AVB
- ✓ Broadcast⇒ SDI, AES

☑ Chip level

- \Rightarrow I²S, I⁸S, PDM, TDM
- ⇒ On the way up: Slimbus





Sometimes all in one device!





Modern audio protocols

☑ Linear PCM

- \Rightarrow Typically a .wav file.
- ⇒ On digital media like CD, DVD where size is no issue
- ⇒ 10 MB per minute... per channel... = huge

☑ Encoded

- ⇒ To save space
- ⇒ To transport multiple channels across systems designed for two channels
- ⇒ Can be "lossy" like MP3, Dolby Digital,
- → Or lossless like Dolby TrueHD, FLAC, Apple Lossless

☑ Test implications of encoded audio

- ⇒ Can't use traditional sine wave tests on encoded audio
- Analyzer must have encoder and decoder or encoded file streaming capability
- ⇒ Can check for "bit accuracy" by comparing to reference file

Dolby & DTS



☑ Many different formats

- ⇒ Dolby Digital, Dolby Digital Plus, Dolby TrueHD
- ⇒ dts-HD Master Audio'

☑ Many different technologies

- ⇒ Receiver, DVD, Blueray, Broadcast, mobile,
- Extremely complicated requirements for certification
 - ⇒ Up to one week to complete all tests
- ☑ Automation can save a lot of time
 - Dolby & DTS provide pre-defined audio analyzer projects

DOLBY.









DTS Certification Process





Why testing is important

How much is an hour of R&D?How much is a recall or returns?



Audio analyzer or sound card?





Audio

Audio analyzer or sound card?

☑ Trust & reliability

- ⇒ Believe your own results
- ⇒ Others believe your results
- ➡ Calibrated & traceable

Is the error you see caused by your product or your "analyzer"?



Audio

Audio analyzer or sound card?

Productivity & sophisticated measurements

- ⇒ No wasted time
- ⇒ Built in measurements & proven algorithms
- ⇒ Easy to share test data with others
- ⇒ Professional technical support

Who do you call when there's a problem?

es	Last Post
В	The project has ended? (chris319) May 14, 2012 04:53 PM
	Beta socks (tkchuong) April 15, 2011 09:15 PM



Audio

Audio analyzer or sound card?

☑ Connectivity & performance

- → Must be able to connect to the device directly
- ⇒ Must have better performance than the device tested



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Do you trust a hacked home receiver to test your new product?



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✓ Sine waves

- ⇒ Single sine wave or stepped sine sweep (level, frequency)
- ⇒ Classic traditional test method, international standard
- ⇒ Ideal for amplifiers, high performance recording devices
- ⇒ Not great for voice-quality devices
- ⇒ Stepped sine sweep can be very slow





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☑ Multitone

- ⇒ Single signal with multiple tones, typically 3,5,7 or 31 tones
- ⇒ Extremely fast: 20+ results in <2 seconds
- ⇒ Ideal for production lines, broadcast, and encoded formats
- ⇒ Some results differ from sine. Not wrong. Just different. TD+N vs. THD+N
- ⇒ Triggering waiting for the right signal





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☑ Continuous sweep (logarithmic sweep or "chirp")

- ⇒ Measurement can calculate Impulse Response (IR)
- ⇒ Results correlate with traditional stepped sine sweep
- Again, extremely fast: 20+ results in <2 seconds</p>
- ⇒ Good for electronic and loudspeaker test, excellent SNR





☑ Ultra high bandwidth measurements

- ⇒ Required for modern Class D amplifiers
- Switching frequencies at 200-300 kHz
- ⇒ Need high bandwidth and high resolution



Calibration



☑ Similar, but very different

- Adjustment
- ⇒ Calibration
- ⇒ Accredited Calibration
- ☑ Traceability
 - ⇒ an unbroken chain of comparisons
 - ⇒ measurement uncertainty
 - ⇒ documentation
 - ⇒ competence
 - ⇒ reference to SI units
 - ⇒ calibration intervals

Audio precis	c SION [®]	alibration Services 5750 SW Arctic Drive Beaverton, Oregon 97005 USA	ACCREDITED Calibratice Cet #2527.01	
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Description: Manufacturer: Model: Serial Number: Customer Name: PO Number:	Audio Analyzer Audio Precision APx525 APX2-00000 Audio Precision	Received Date: Date of Calibration: Procedure: Report Form: Control Software: Temperature 21°C: Relative Hurddhy: AC Mains: Data Type:	Sew Internegat 28-Jun 2017 APrCalifornian get 2.77 APs25CalReport.xls	
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		Page 1 of 7		





Calibration





Common mistakes: grounding

- ✓ Most common mistake we see
- ✓ Star grounding is best
- ☑ Use heavy gauge and always check your grounds



The resistance in each leg of the chain puts the devices at different ground potentials, and is not as effective as star grounding.



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Common mistakes: Class D filtering

- ☑ Class D switching amplifiers have unique problem of out of band noise
- ☑ Can cause inaccurate measurements, or even damage analyzer inputs
- ☑ Simple filter stops the problem





Filter effect in Frequency Domain





Typical Class D
 Low-level
 Output Signal
 Audio signal is masked
 by switching artifacts.



Audio

Output Signal after PREanalyzer filtering Audio signal is now clearly visible.









Common mistakes: Connections

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- ✓ Sounds simple, but it happens a lot
- ☑ Crossed cables
- ☑ Bad cables
- ☑ Analyzer set for wrong connections
- ☑ Always check your connections
- ☑ Use Loopback mode to confirm settings





Demo and questions





Audio analyzer comparison







2700 Series

ab.com

The world's highest performance audio analyzer

Most advanced analyzer for R&D

- Vanishingly low residual noise and THD+N: —115 dB @ 2.0 Vrms
- True Dual Domain analog and digital
- Generate and analyze a wide range of waveforms
- API automation
- LabVIEW integration
- Chip-level connectivity with PSIA
- User-defined sweeps, switcher support up to 192 channels



THD+N of AP 2700 (Black) compared with 5 other audio analyzers





APx Series

Connectivity, Flexibility and Intuitive Operation

High performance for R&D

- Up to -110 dB THD+N
- Test Bluetooth, HDMI, 12S
- 1 million point FFT analyzer with 24 bit resolution from DC to 1MHz
- Multichannel and high bandwidth options

Multiple interfaces

- HDMI, Bluetooth, PDM
- AES/EBU, I²S, S/PDIF

Intuitive & easy to use

- One-click measurements
- Real-time monitors
- Automated reports

Fast, simple automation for production

- Up to 21 measurements in 1.2 seconds
- Automation without coding or use the .NET API and LabVIEW driver
- Lockable projects with limits









APx Series Family

APx525 family: Audio testing for modern devices





- Typical THD+N better than –I 10 dB
- IM point, I MHz FFT w/BW52 option
- 2 or 4 independent analog inputs, 192k digital I/O, digital serial I/O
- Maximum input voltage 300Vrms (bal)
- Regulated frequency sweeps, CMRR, CEA-2006 / CEA-490A measurements

APx525 2 channel analog APx526 4 channel analog



- 8 or 16 independent analog inputs 192k digital I/O, Digital Serial I/O
- Systematic test of HDMI & Blu-ray
- Automated Dolby/DTS compliance test with fully formatted reports
- Test EDID error handling with metadata event timing view

APx585 8 channel analog APx586 16 channel analog



- Typical THD+N better than –106 dB with a production test price point
- Comprehensive automated test in 3 seconds without any coding
- Small footprint, only 4 kg
- Share tests and results with any APx seamlessly
- APx515 2 channel analog + 192k digital

