

## Audio Component measuring system

# Audio-Capture

The Audio Capture is a PC based multifunction audio testing system that is ideal for designing, verification and quality check of audio components such as loudspeakers, acoustics and audio electronics.

Audio-Capture acquires complex frequency response by applying either a maximum-length sequence (MLS) stimulus, Sweep tone (time-domain chirp), Gated Sine or Multitone to the unit or acoustic environment under test.

Audio-Capture interfaces to a quality stereo or four to eight channel soundcard. Simultaneous measurements can be made on multi-channel units or in multiple positions.

For production quality tests or performance check a Quality Check module can perform Pass / Fail judgement of up to three parameters at the same time. A user-friendly graphical interface creates the tolerance limits.

For the loudspeaker designer, loudspeaker boxes can be simulated by either a database of Thiele-Small parameters or acquired by impedance measurement. Active and passive crossover networks can be simulated. Polar- and

Directivity-plots can be acquired. Audio-capture supports several industry standard turntables.

For bench test of electronics and sound reinforcement distortion test, Audio-Capture offers several software instruments such as; Spectrum analyser, Dual channel oscilloscope, AC Volt and Watt meter, IM-Distortion analyser, Function generator and LCR-meter.

### Benefits

- Multiple measurements can be taken at several measurement locations and averaged.
- Quality check with tolerance limits for Pass / Fail with post-processing reporting.
- Fast and reliable measurement using MLS stimulus using up to 4 millions points.
- Accelerated Log Sine Sweep (Chirp) increases the dynamic measured range at low frequencies.
- Measurements can be presented as Waterfall graphs or Polar plots.
- Loudspeaker box and crossover simulation using impedance measurement and Thiele-Small parameters.

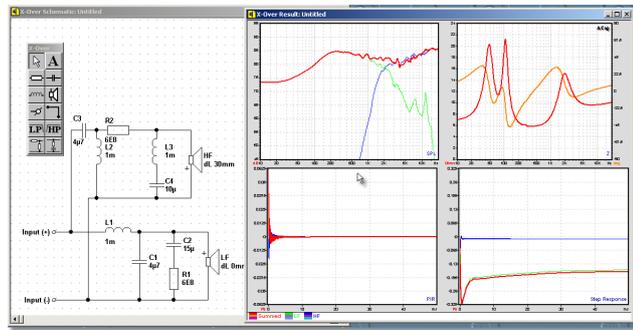
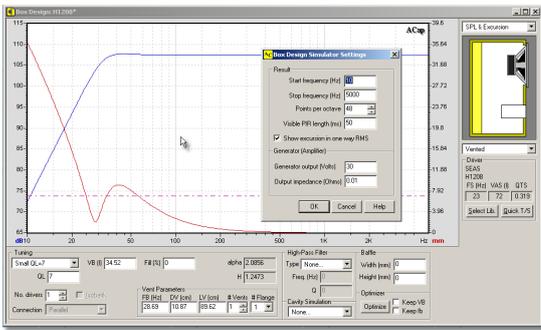
### Features

#### Time-Domain Stimulus Types

- White MLS - Length: 255 – 4.194.303 points for all MLS types
- Pink MLS
- Chirp - Log Sine Sweep that is converted to time-domain
- Spike - The oldest method for impulse measurement
- Sin2 pulse - Squared sine pulse
- External Wave File - One shot Dual Channel FFT with music as stimulus

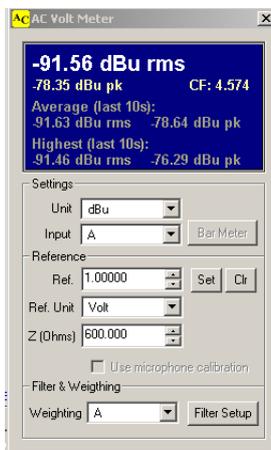
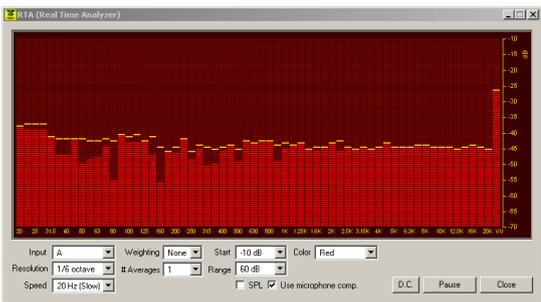
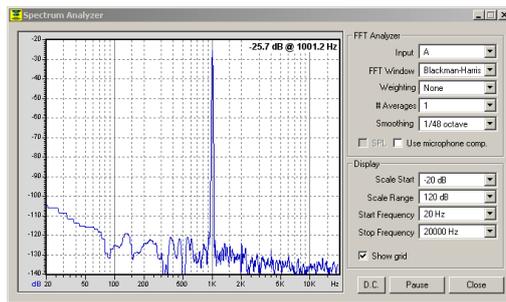
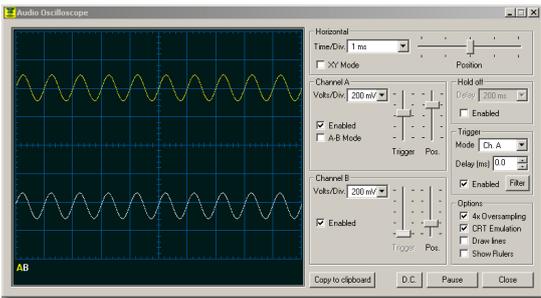
#### Frequency-Domain Stimulus Types

- Gated Sine Wave - Sine bursts used for distortion measurements
- Log Sine Sweep - Sweep time up to 10s per octave and a maximum resolution of 4096 ppo
- Multitone - Ideal for Rub'n Buzz measurements in production
- Amplitude Sweep - Sweeps Level, Power or THD vs. amplitude



### Analysis Result Windows

- Frequency Response** - Industry standard windowing and convenient operations such as Smoothing, Normalize, Average etc.
- Polar-Plot** - This window is combined with a frequency graph where you can select which frequencies that is visible in the polar-plot. Support for industry standard turntables. It also has directivity-plot.
- Impedance** - This can show both single impedance curves and complete Thiele-Small impedance curve pairs. A Nyquist plot can also be shown.
- Impulse (PIR)** - This window also have these result curves: Step Response, Shroeder Plot, ETC, Log Square and Cepstrum.
- CSD (Waterfall)** - This can be shown in standard 3D form or as a sonogram where the colour show the magnitude.
- Multi-Graph** - Can show several frequency and/or impedance curves at the same time (good for comparison and documentation).



### Instruments

- Mille-volt and dB meter with RMS, Peak and Average detectors, weighting and band-pass filters
- Distortion Analyser; THD, IMD, SMPTE, CCIF and DIM
- L-C-R meter
- Function Generator; with sine, triangle, saw tooth, square, white/pink-noise and bursts waves
- Oscilloscope, with trigger filtering and phosphor emulation
- Spectrum Analyser
- RTA ; with 1, 1/3, 1/6 octave resolution and weighting filters
- Microphone Calibrator; in conjunction with a calibrated microphone is it very easy to calibrate an unknown microphone

### Requirements

- PC with Win 2000 SP4 or XP
- Soundcard with 2 inputs and 2 outputs (duplex mode) or up to 8 channels
- For loudspeaker equalization:
  - A flat measurement microphone and pre-amp
  - Loudspeaker power amplification
  - Digital/analog equalizer or loudspeaker digital signal processor

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