



# **Case Study:**

## **Evaluating a Laptop's Audio Performance**

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# Abstract

The performance of a laptop computer's audio system is evaluated using the eightolives AudioAnalyzer.

# Background

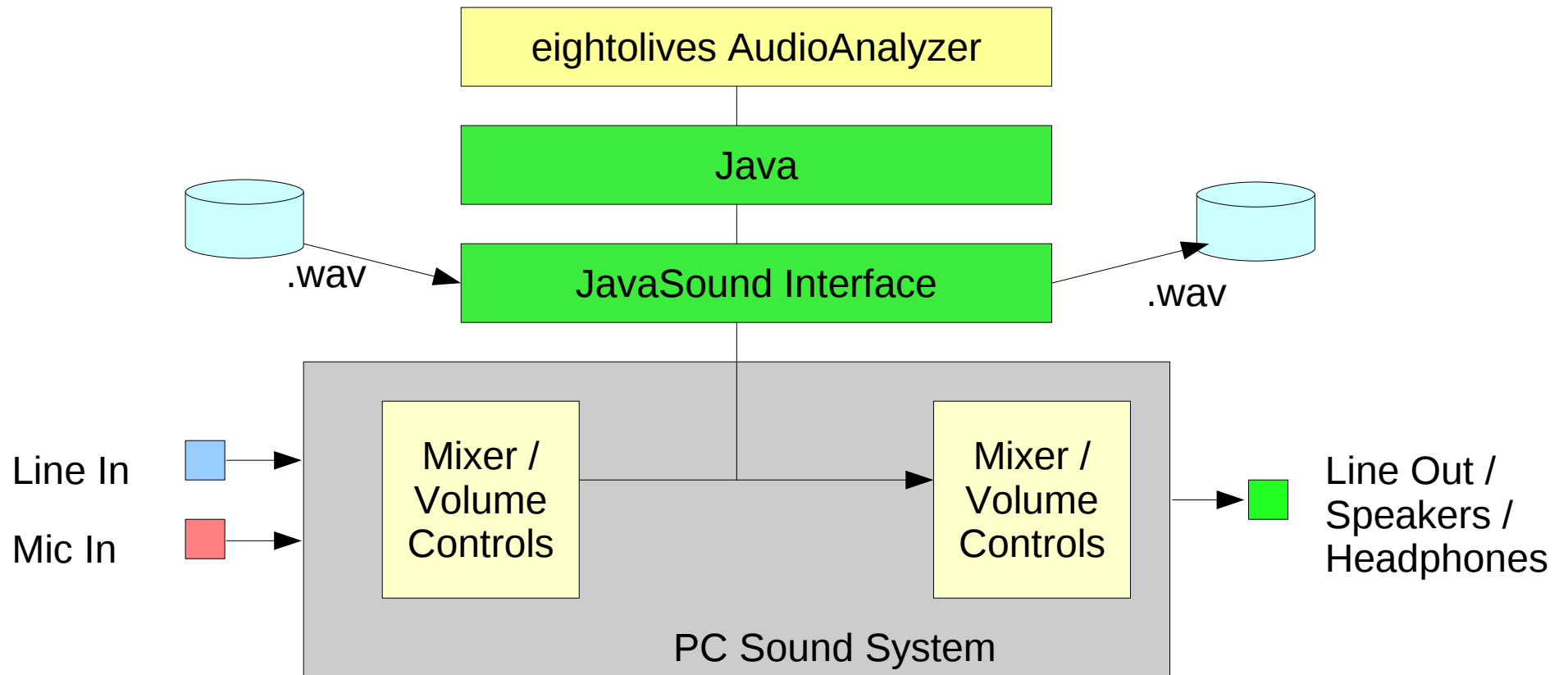
- The eightolives AudioAnalyzer is a Java-based program that provides a set of audio tools that can interface to a PC's audio system (uses the Jwasound interface)
  - 2 – Waveform Generators – sine, triangle, rectangle, white noise, AM, FM, phasing, harmonics
  - Oscilloscope for viewing 2 channels
  - Spectrum Analyzer – FFT view of the audio spectrum
- Objective is to see how well it performs on a typical PC

AudioAnalyzer = Waveform Generator + Oscilloscope + Spectrum Analyzer + Recorder

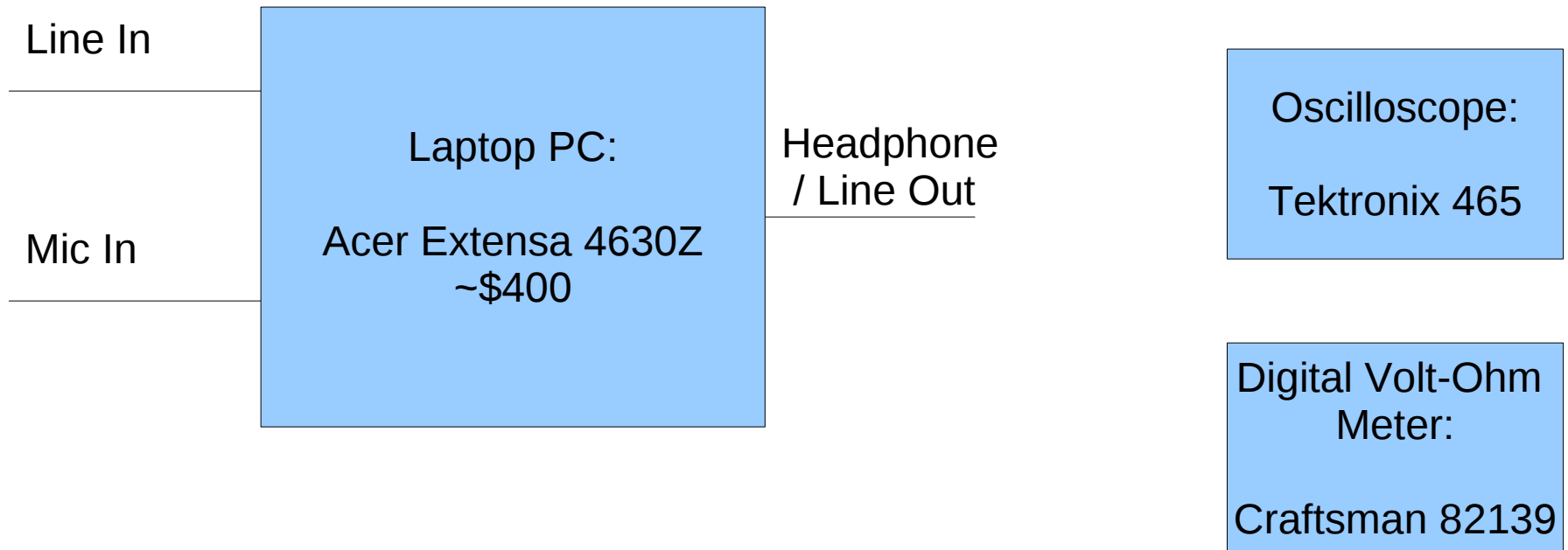
The screenshot displays the 'eightolives AudioAnalyzer' software interface, which is divided into several functional windows:

- Waveform Generator:** This window is split into two channels. Channel 1 is configured with a Sinewave waveform at a frequency of 1000 Hz and an amplitude of 20 dB. Channel 2 is also configured with a Sinewave waveform, but at a frequency of 1892 Hz and an amplitude of 0 dB. A phase lag of 90 degrees is specified for Channel 2. Both channels have a frequency range of 20 - 20000 Hz and a time range of 0 - 20000 us. 'Start' and 'Stop' buttons are located at the bottom of this window.
- Spectrum Analyzer:** This window displays a frequency spectrum plot with a prominent peak at 1892 Hz, corresponding to the signal in Channel 2.
- Oscilloscope:** This window shows two waveforms: a high-amplitude yellow sine wave for Channel 1 and a lower-amplitude green sine wave for Channel 2. Two vertical cursors are placed on the waveforms. The status bar at the bottom indicates 'Running'.
- Audio Center:** This window is located in the bottom-left corner and features a 'File Select' button, a 'Record Enable' checkbox (which is currently unchecked), and 'Record', 'Play', and 'Stop' buttons.

# The Sound System



# Test Setup



Both Windows XP and Linux were used in the tests on the same hardware.

# Interface Basics

- Line In – 3.5 mm (1/8”) Tip Ring Sleeve (TRS) connector (light blue)
  - Nominal level = .316 Vrms (.447 V peak, .894 Vpp)
  - Input impedance ~ 10 Kohms
- Line Out – 3.5 mm TRS (lime green)
  - Max output voltage ~ 2 V peak-peak
  - Output impedance ~ 100 ohms
  - Frequency Range = 20 Hz – 20,000 Hz

Source: Wikipedia

# About Signal Levels

- Waveform Generator defines the maximum undistorted sine wave amplitude as +20 db
  - For 16 bit resolution, +20 db peak-to-peak is digitally expressed as +32767 to -32768
  - 0 db (nominal level) peak-to-peak is 10 times less
- Actual analog output levels on Line Out are determined by the PC mixer's volume controls
- Analog input signal levels are also determined by the PC mixer's volume controls



# Javasound Interface

- Javasound supports reading and writing .wav files
- Supports sample rates of 8000, 11025, 16000, 22050, 44100 Hz
- Supports 2 channels (stereo)
- Supports 8 bit and 16 bit resolution

# Observations on Same PC

- Using Windows XP
  - Max output on LINE OUT was 4.8 Vp-p
  - LINE IN had no attenuation
- Using Linux
  - Max output on LINE OUT was 2.8 Vp-p
  - LINE IN has a -20 db attenuation to the AudioAnalyzer

# Observations (Linux)

- With audio system volume controls at max and AudioAnalyzer's volume control at 20 db (max)
  - Output sinewave was 2.8 Volts peak to peak (Vpp)
  - Response was the same 20 Hz to 20 Khz
  - At 1 Hz output was 1.4 Vpp
  - At 22050 Hz output was 1.2 Vpp
  - Channel 2 was about 0.1 V less than Channel 1

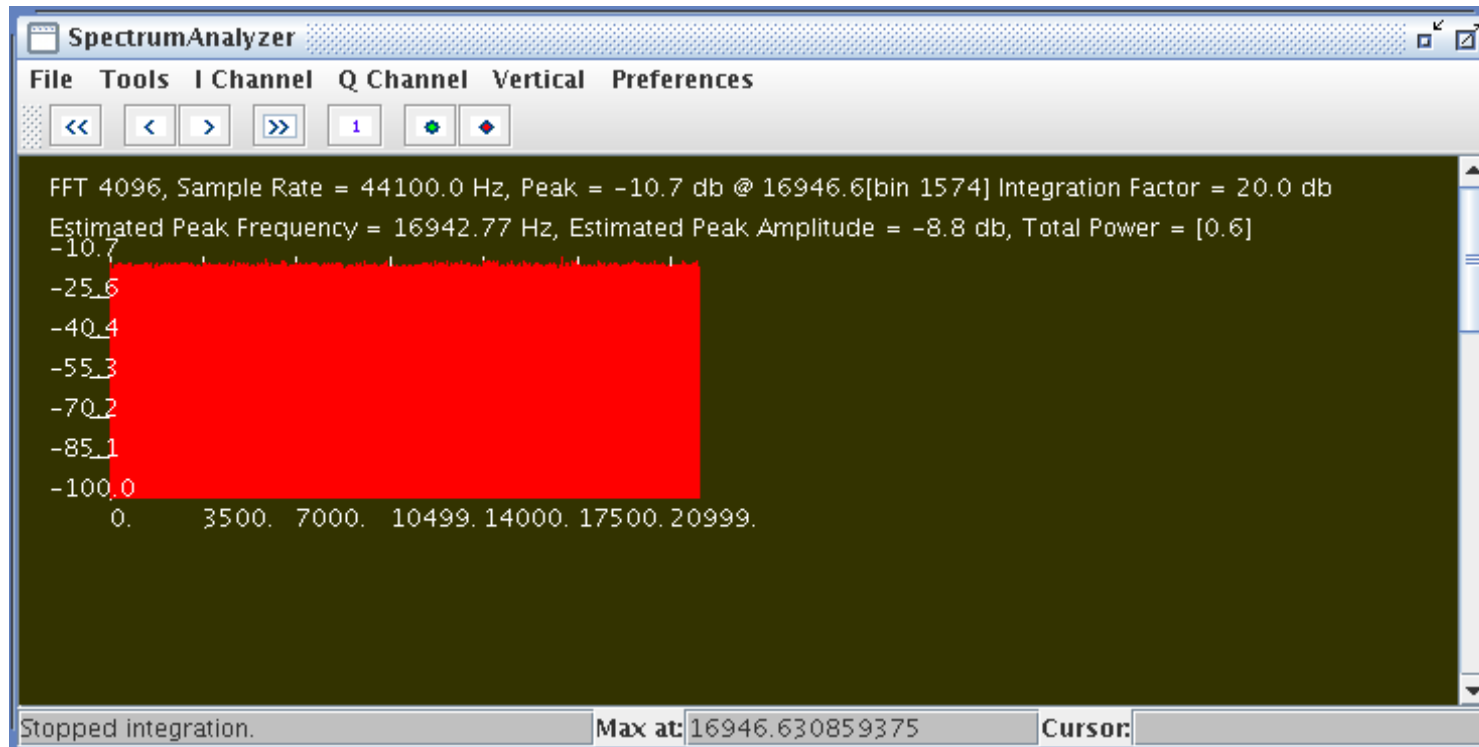
# LINE OUT to LINE IN (Linux)

- Connecting LINE OUT to LINE IN
  - Observed amplitude from Mixer displayed on AudioAnalyzer's Oscilloscope was 0.134 V<sub>peak</sub> vs 1.4 V applied to the input or – 20db
  - At 20 Hz V<sub>peak</sub> = 0.187 V
  - At 10 Khz V<sub>peak</sub> = 0.134 V
  - The signal on LINE OUT was not affected by connecting it to LINE IN

# LINE OUT to MIC IN (Linux)

- When LINE OUT was connected to MIC IN, the observed output from LINE OUT was oscillating a squarewave with an amplitude of 1.2 V peak to peak at 800 Hz

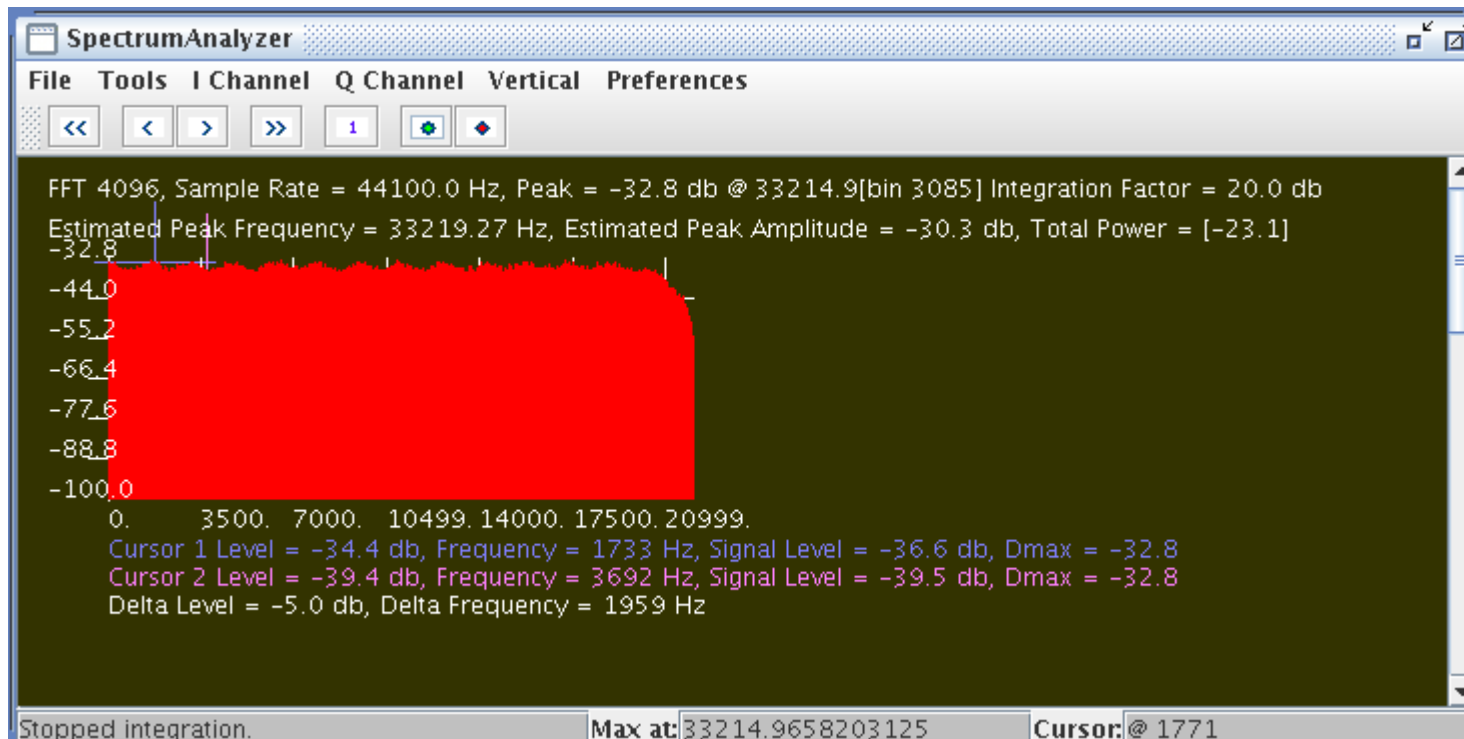
# Using White Noise



The White Noise from the Waveform Generators produce a flat spectrum.

The display shows 10 passes integrated.

# Connecting LINE OUT to LINE IN



Looking at the Mixer input with LINE OUT connected to LINE IN shows the filtering of PC's output and input stages.

The 10 integrated passes show +/- 2.5 db ripple in the bandpass, a roll-off at 19 KHz with response being -30 db lower at 22050 Hz, and a -20 db overall loop gain.

# Other Observations

- The polarity of the output signals was inverted from that displayed on the AudioAnalyzer's Oscilloscope.
- An external speaker system connected to the Headphone jack had no output below 200 Hz or above 7 KHz
  - Same observer using headphones:
    - 20 to 8700 Hz



# For more information

- Check the tutorials at:  
<http://www.eightolives.com/tutorials.htm>