

# DSP4Linguists

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July 30, 2021

# Acoustic wave

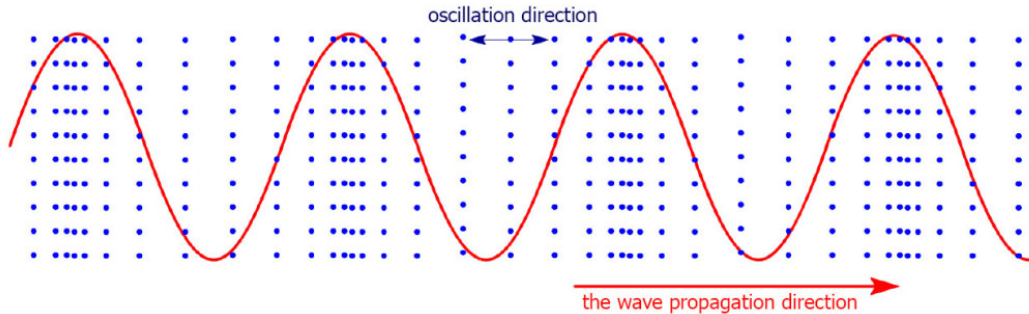


Figure 1: Acoustic wave.

## Analog > Digital / Digital > Analog

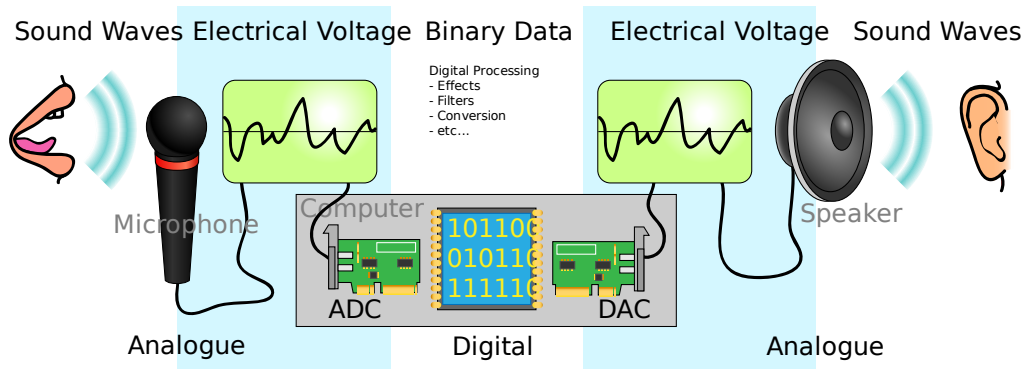


Figure 2: ADC and DAC.

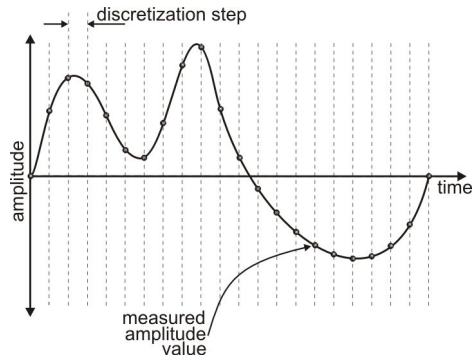


Figure 3: Sampling.

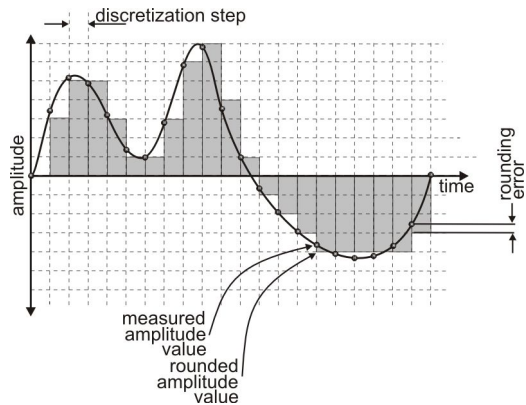


Figure 4: Quantization.

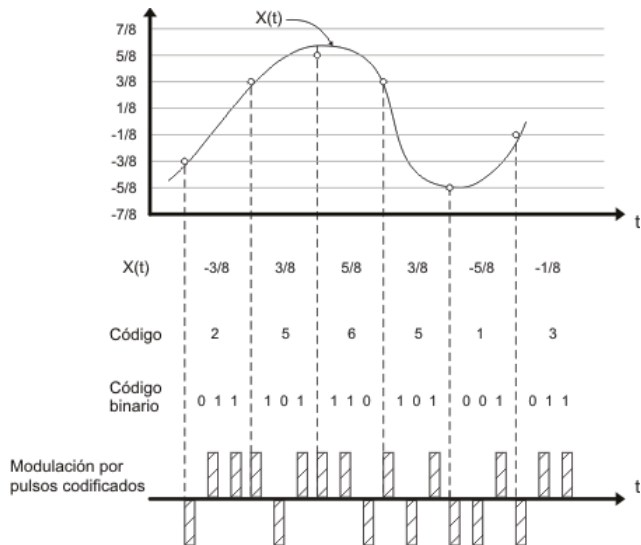
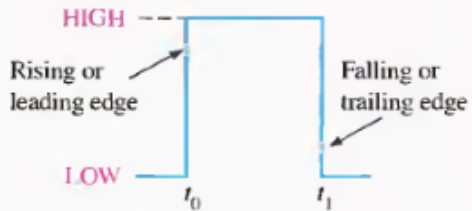
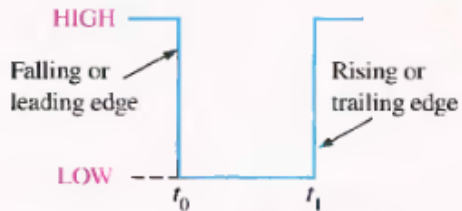


Figure 5: Coded pulses.



(a) Positive-going pulse



(b) Negative-going pulse

Figure 6: Digital pulse.

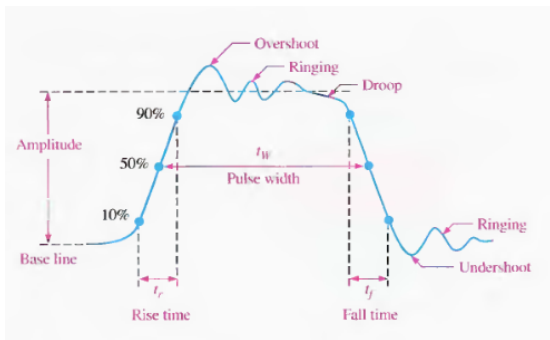


Figure 7: Non-ideal Pulse.



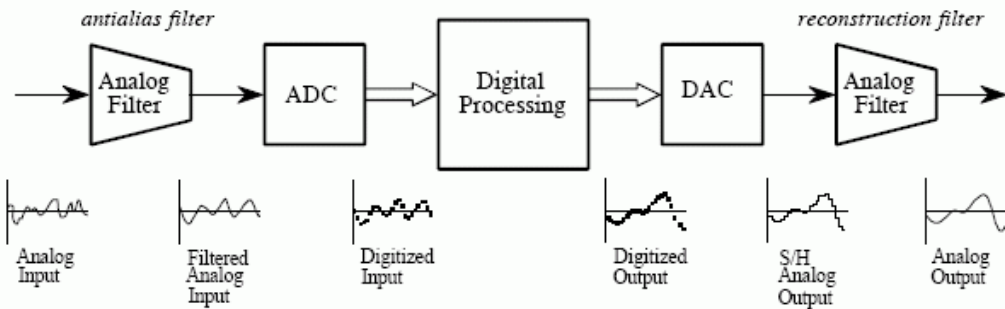


Figure 8: ADC and DAC

# Quantizer

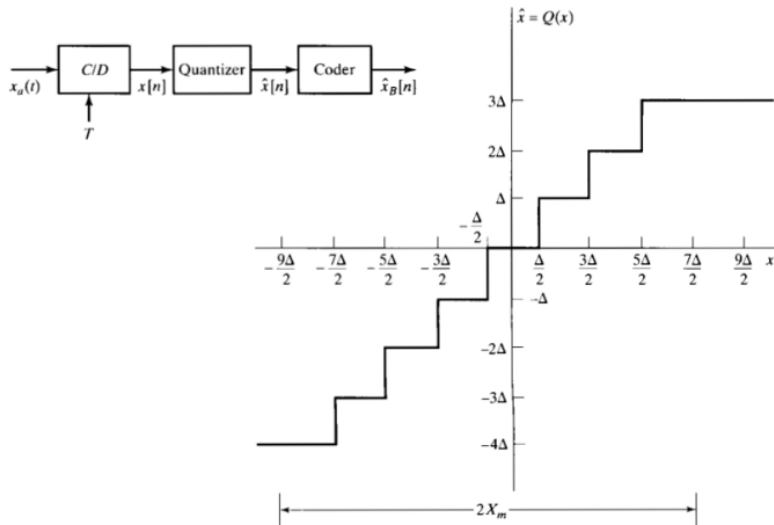


Figure 9: 3 bits uniform quantizer

## Quantization examples

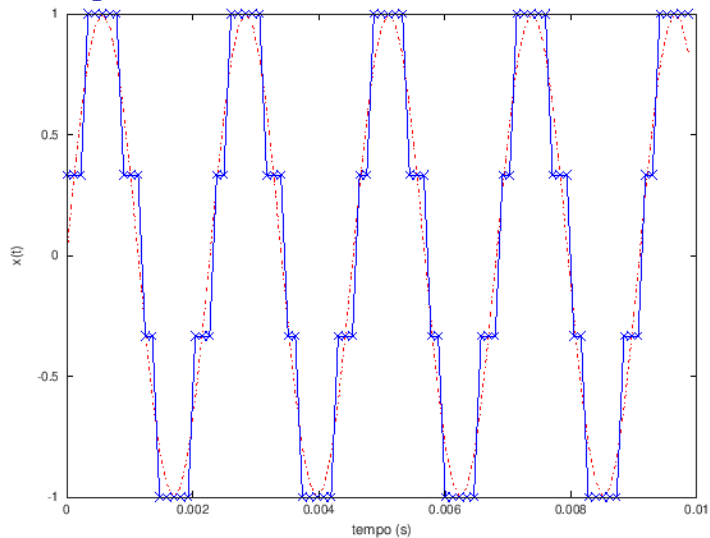


Figure 10: 440 Hz sin wave.

## Clipping

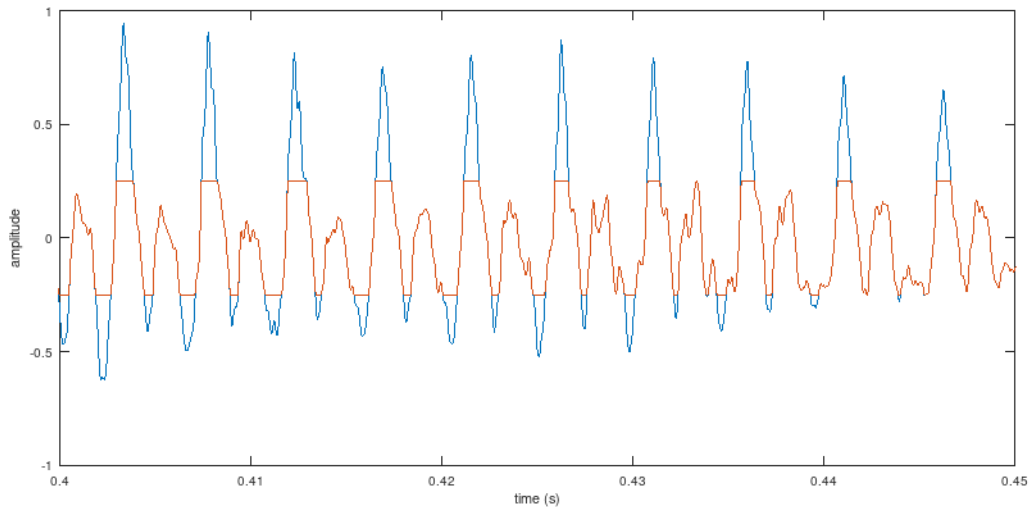
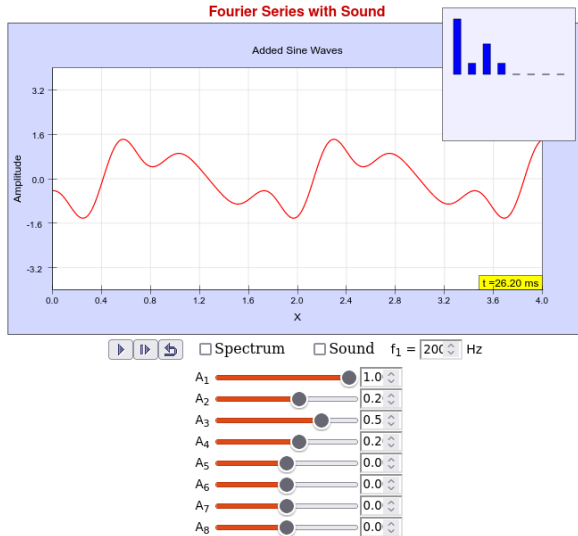


Figure 11: Signal clipping

# Fourier Series

## Simulation



# Signals and Systems

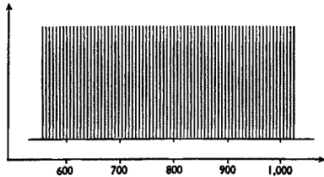


Fig. 5.3. The spectrum of a sound consisting of a large number of tones with the same amplitude.

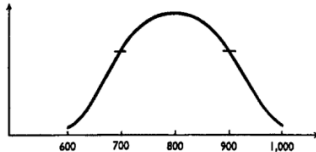


Fig. 5.4. A curve specifying a resonator.

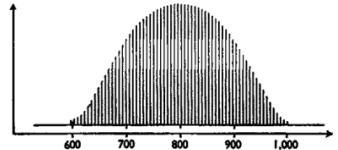
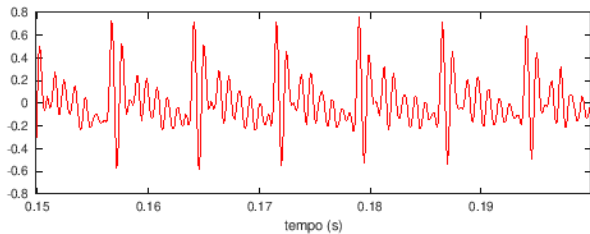
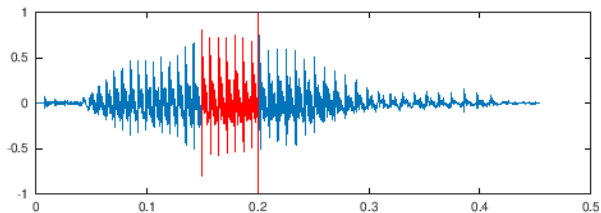


Fig. 5.5. The output of the resonator in fig. 5.4 when the input shown in fig. 5.3 is applied to it.

Figure 13: Resonator

# Vowel

open\_front\_unrounded.mp3



## Pitch

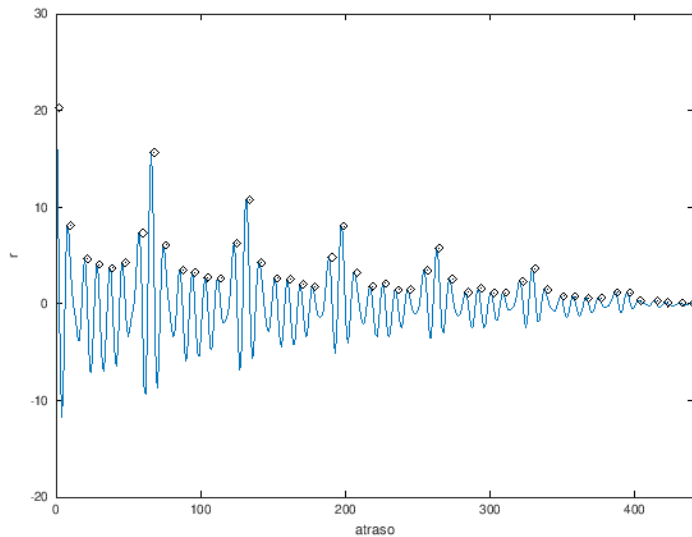


Figure 15: Autocorrelation



# LPC model

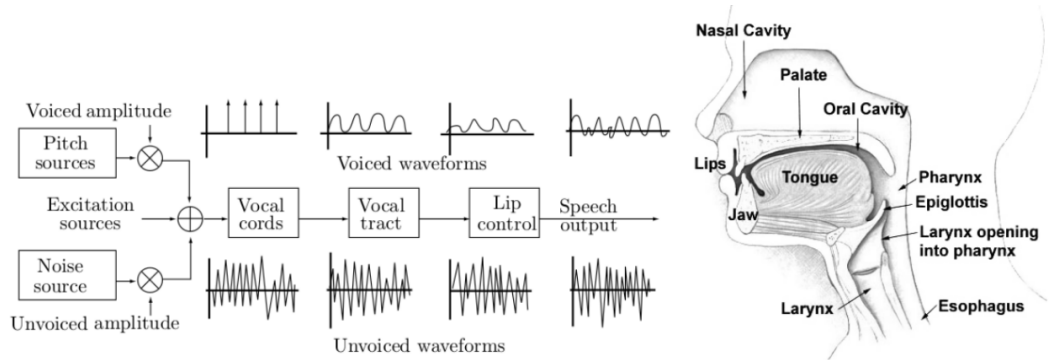


Figure 16: LPC model and vocal tract

## Vocal apparatus

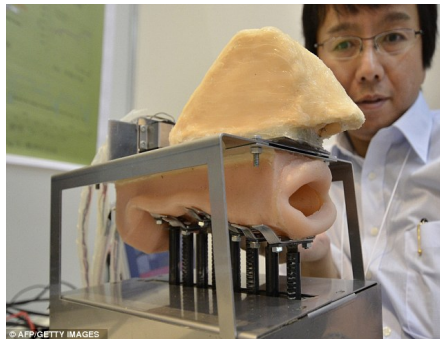


Figure 17: Hideyuki Sawada's KTR voice robot  
[https://www.youtube.com/watch?v=qobhDJ\\_vEOc](https://www.youtube.com/watch?v=qobhDJ_vEOc)

vocal cord stroboscopy examination

History of Speech Synthesis

four people sing Kyrie eleison during laryngoscopy

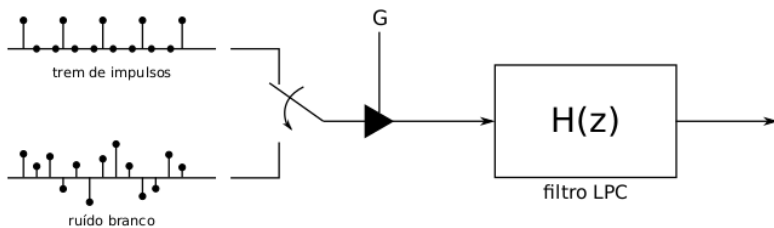


Figure 18: LPC model

synthesized example

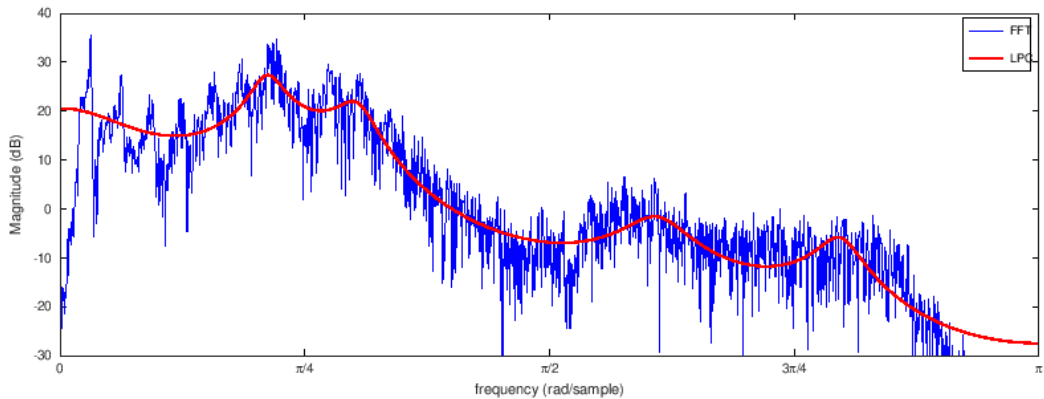


Figure 19: Spectrum and LPC.

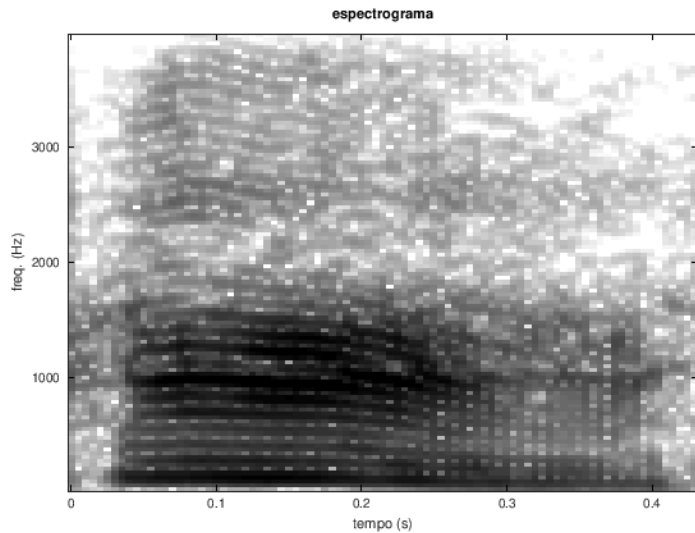


Figure 20: Spectrogram.

# Spectrogram

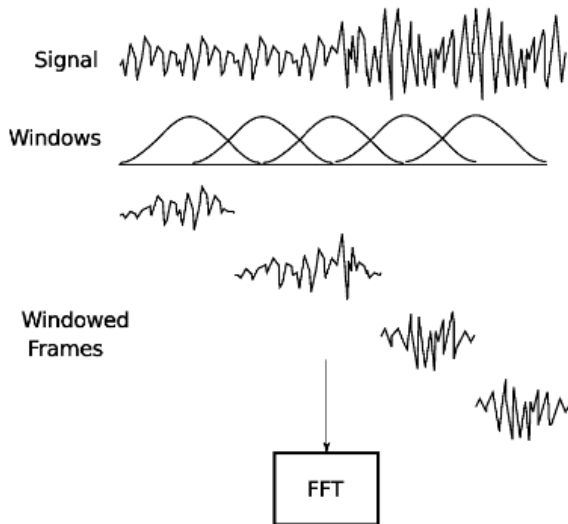


Figure 21: Schematics

## Time vs Frequency

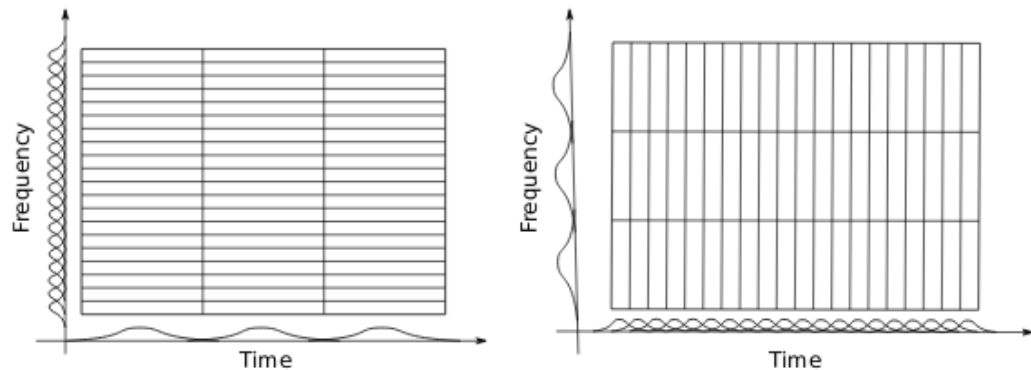


Figure 22: Uncertainty principle

# The Uncertainty Principle



Figure 23: Heisenberg's uncertainty principle



## Spectrogram

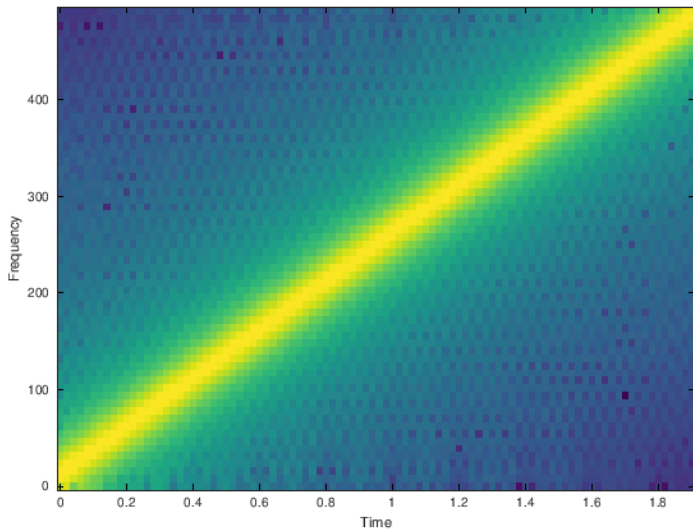


Figure 24: Chirp example, from 0 to 500Hz.

## Shepard Tone

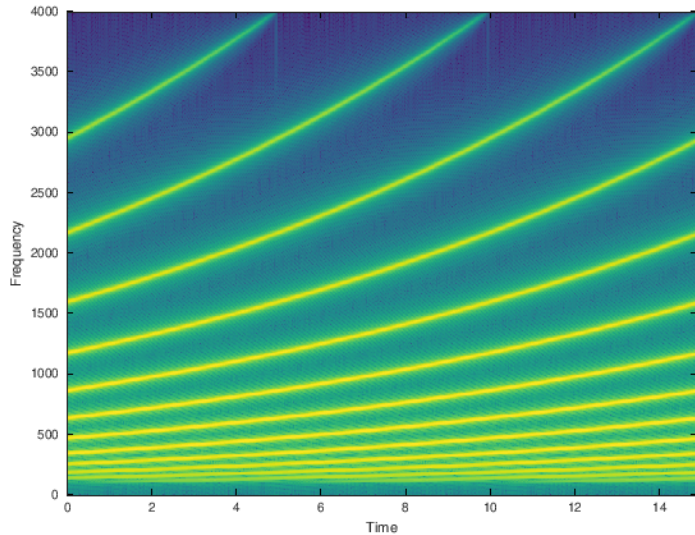


Figure 25: Shepard Tone

## Downsampling

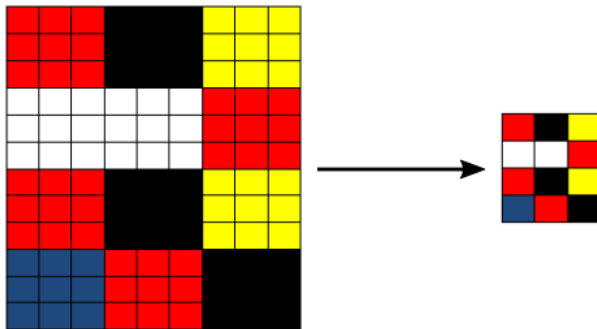
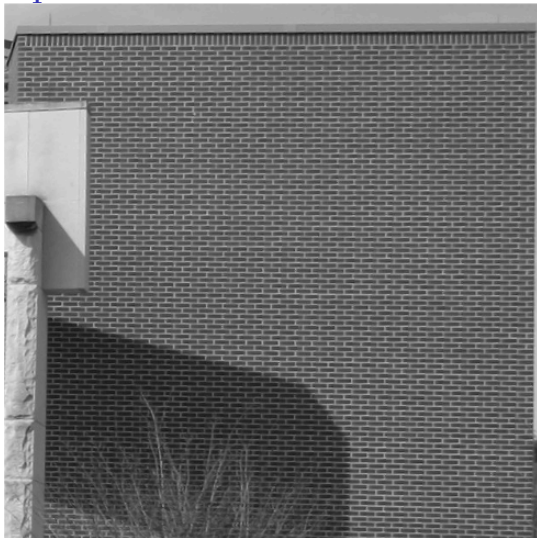


Figure 26: Downsampling example

## Downsample / Decimate



$$\begin{array}{ccccc} x(n) & \longrightarrow & \boxed{\downarrow M} & \longrightarrow & y(m) = x(mM) \\ f_i & & & & f_o = f_i/M \end{array}$$

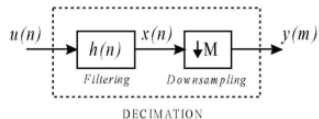


Figure 27: Downsample and Decimate

## Downsample / Decimate (audio example)

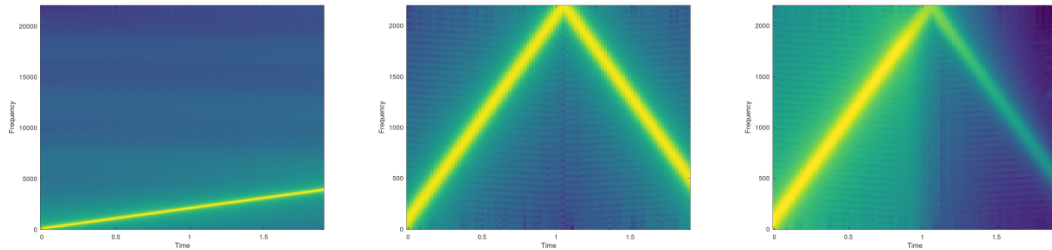


Figure 28: Downsample and Decimate

# Hearing

- ▶ Intensity
- ▶ Pitch
- ▶ Duration
- ▶ Quality/timber

# Pitch metamery

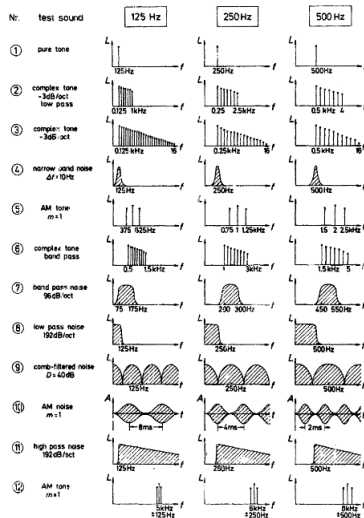
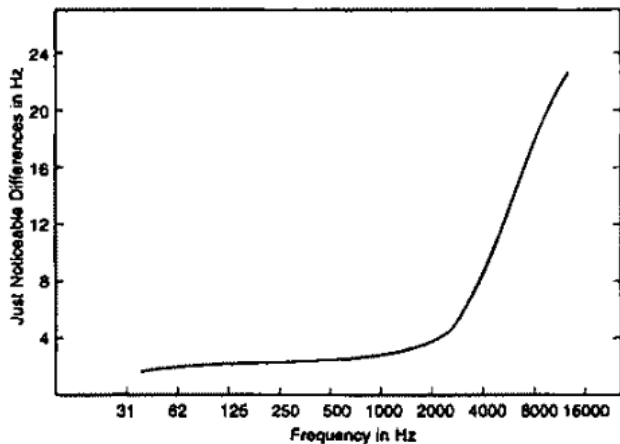


Fig. 1. Schematic representation of test sounds employed.

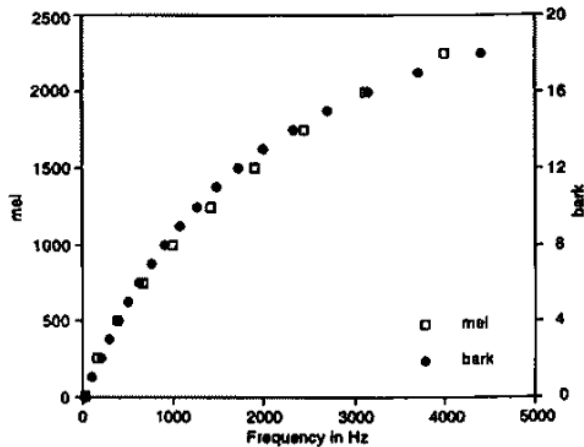
## Pitch JND



**Fig. 6.3. A graph showing how much the frequency of a tone has to be altered in order to produce a change in pitch.**



## Pitch scales



**Fig. 6.4. Graph showing the relation between frequencies in Hz and the corresponding values on two different pitch scales, mel and bark.**

# Duration

- ▶ Staat /ʃta:t/ - country; state
- ▶ Stadt /ʃtat/ - city; town

# Hearing range

esp

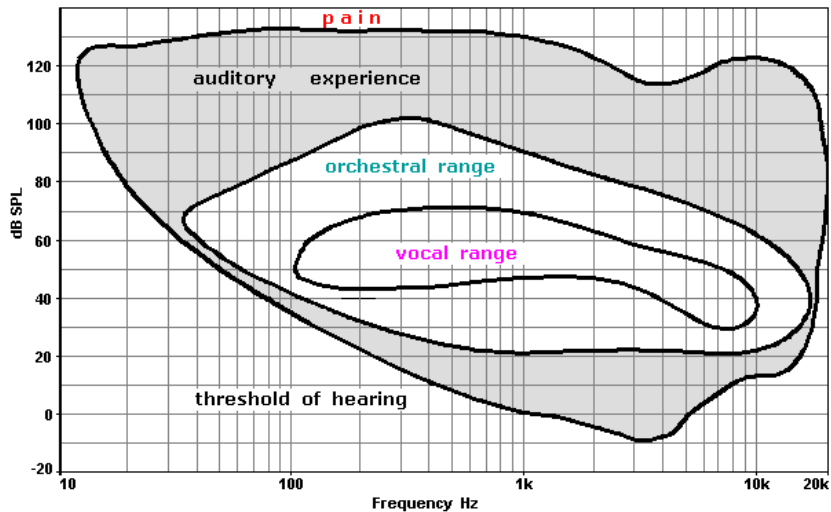


Figure 32: Hearing range

## Fletcher-Munson curves

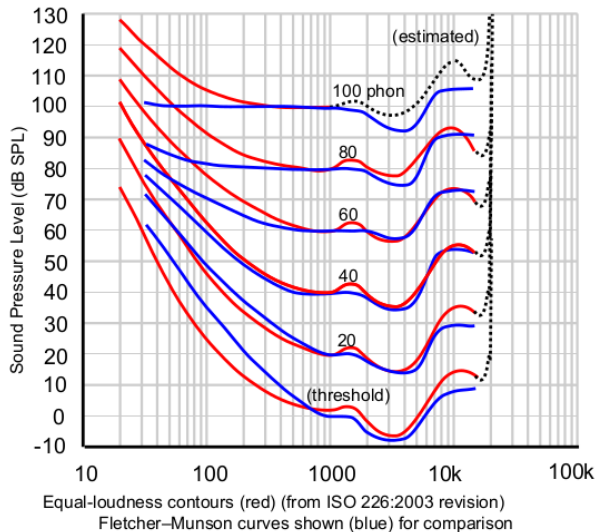


Figure 33: Equal-loudness contour

# Masking

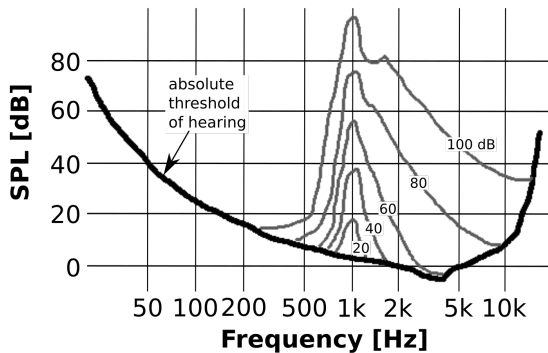


Figure 34: Frequency masking

## Masking

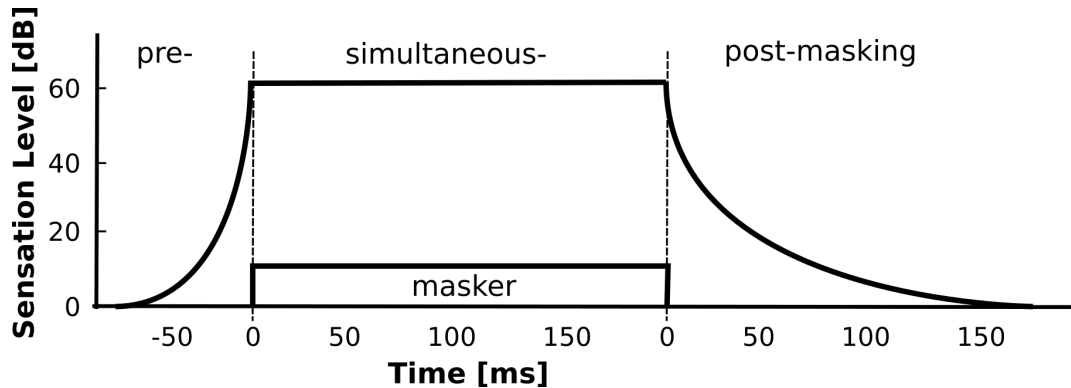


Figure 35: Time masking

# Musical instruments

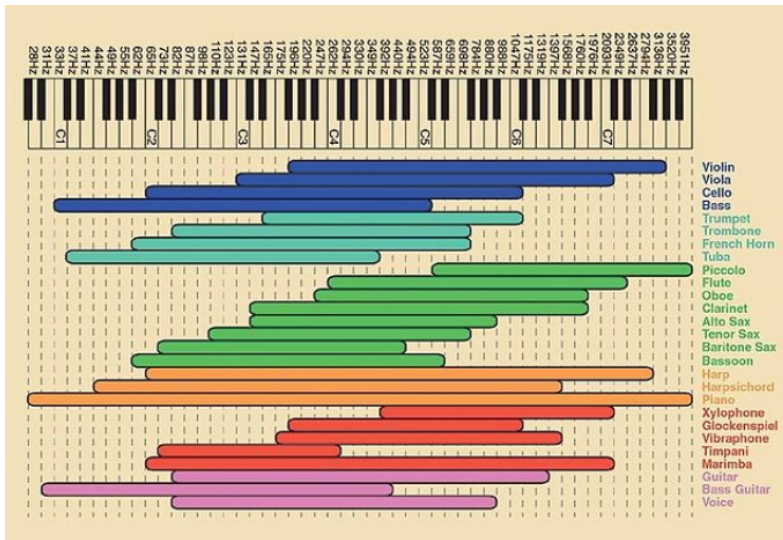


Figure 36: Frequency range of musical instruments

# Ear

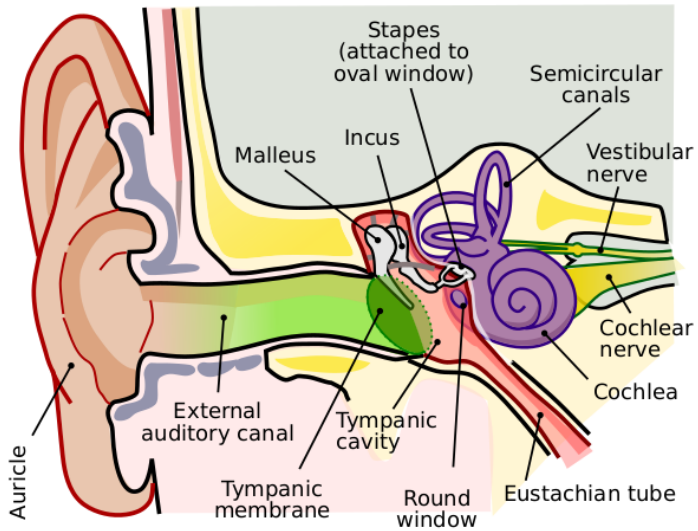


Figure 37: Ear



## Middle ear

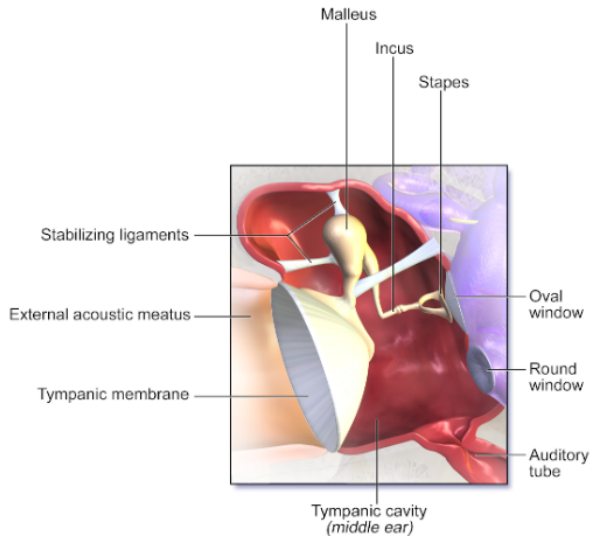


Figure 38: Middle ear

# Cochlea

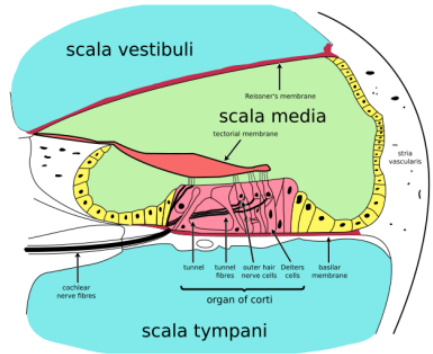
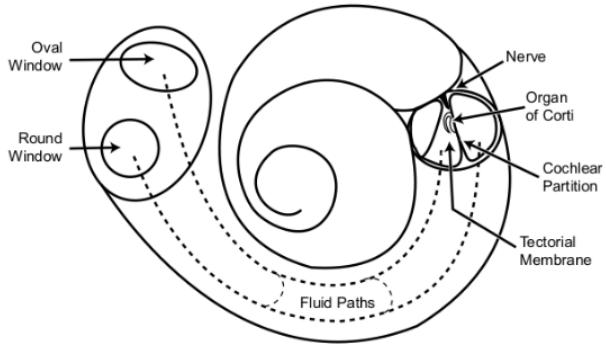


Figure 39: Cochlea and organ of Corti

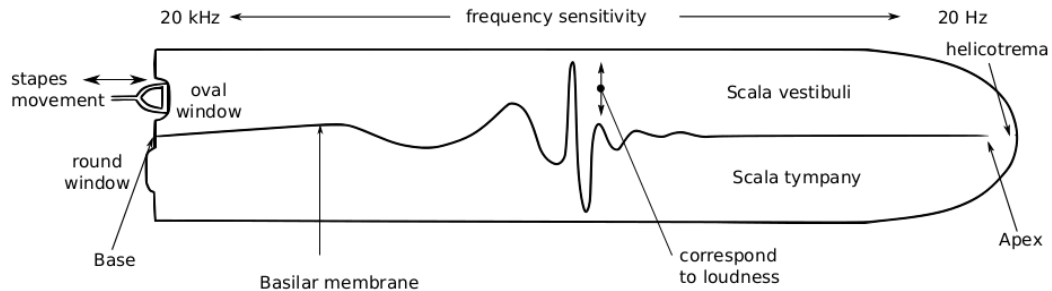


Figure 40: Travelling wave

## Hair cells

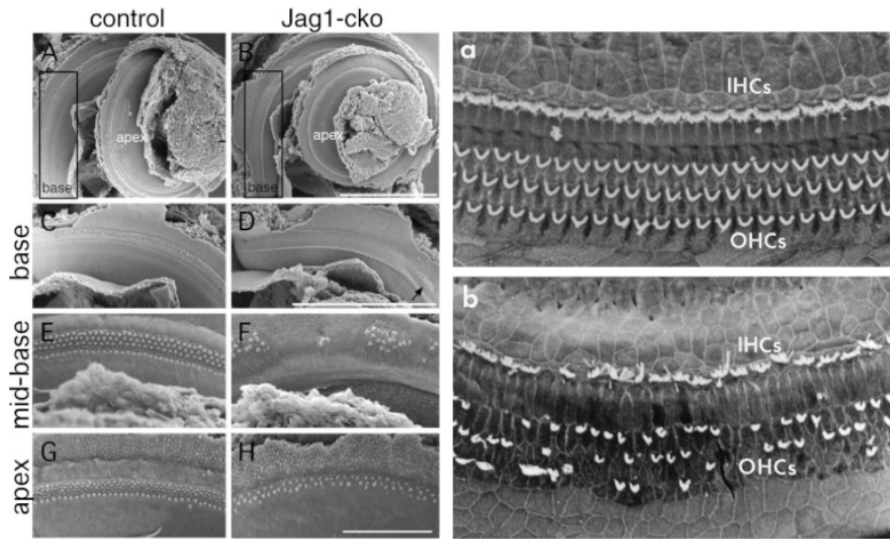


Figure 41: Inner (IHC) and outer hair cells (OHC)

## Auditory Scene Analysis - Albert Bregman

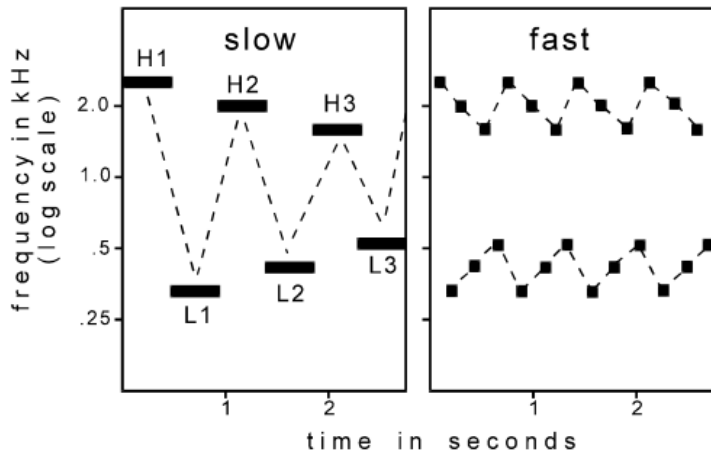


Figure 42: Stream segregation in a cycle of six tones

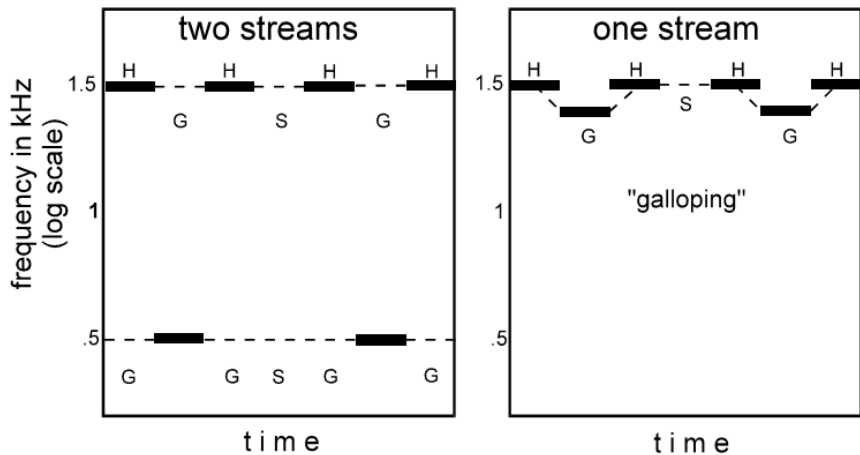


Figure 43: Loss of rhythmic information as a result of stream segregation

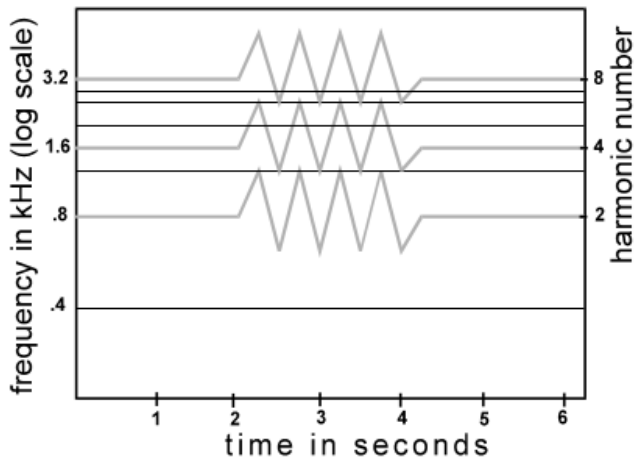


Figure 44: Fusion by common frequency change

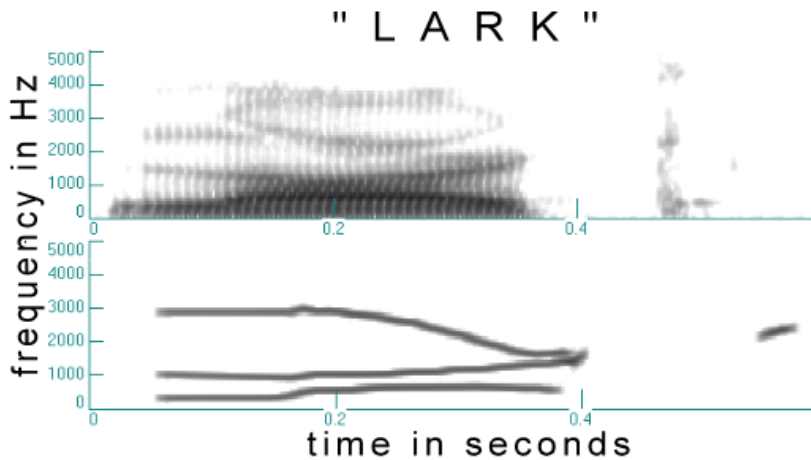


Figure 45: Sine-wave speech





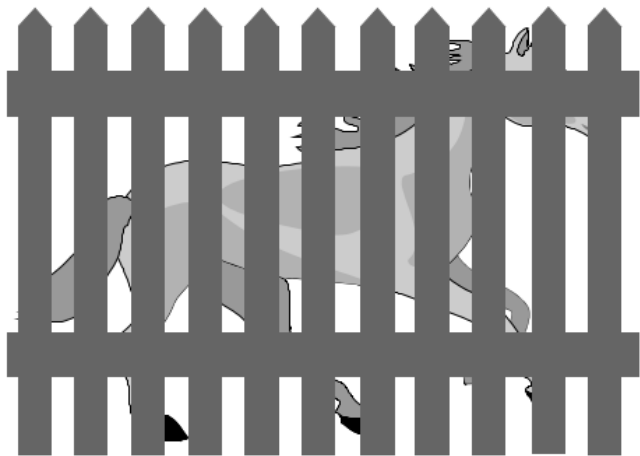


Figure 47: The picket-fence effect with speech

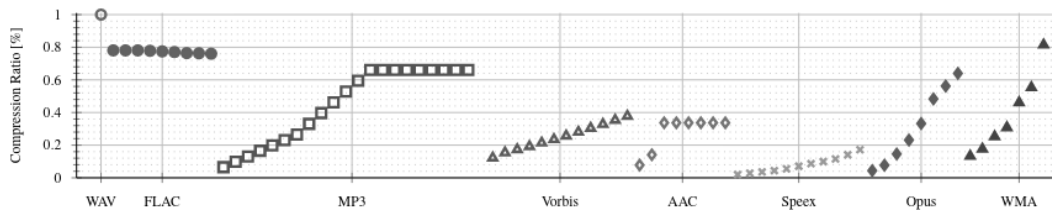
# Audio file formats

**Table 1** - Overview of selected Audio Codecs

Name	WAV	FLAC	MP3	Vorbis	AAC	Speex	Opus	WMA
Released	1991	2001	1993	2000	1997	2003	2012	1999
Compression	No	Yes	Yes	Yes	Yes	Yes	Yes	Yes
Loss-less	–	Yes	No	No	No	No	No	No
Bit-rate (kbit/s)	1,411.2	935	16–320	48–500	16–320	2–24	8–128	32–448
Encoder	–	flac	lame	oggenc	ffmpeg	speexenc	opusenc	ffmpeg
Decoder	–	ffmpeg	lame	oggdec	ffmpeg	speexdec	opusdec	ffmpeg

Figure 48: I Siegert, AF Lotz, LL Duong, A Wendemuth, Measuring the impact of audio compression on the spectral quality of speech data, 2016

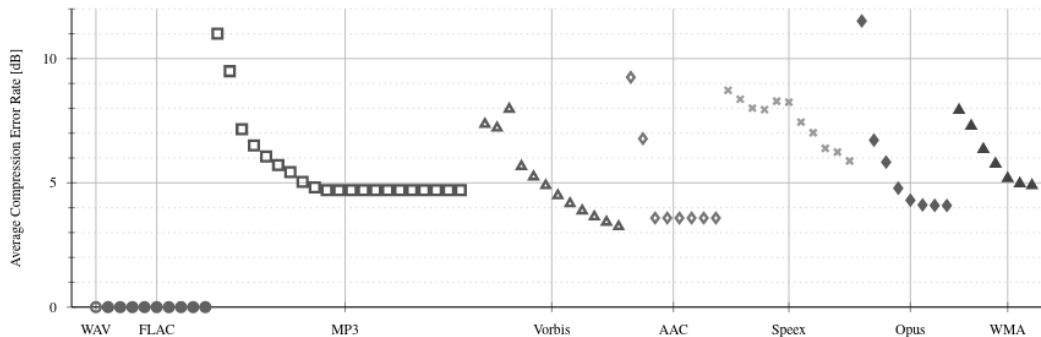
# Compression ratio



**Figure 1** - Achieved average compression ratio for each codec and bit-rate. The bit-rate is increasing from left to right, see Table 2.

Figure 49: I Siegert, AF Lotz, LL Duong, A Wendemuth, Measuring the impact of audio compression on the spectral quality of speech data, 2016

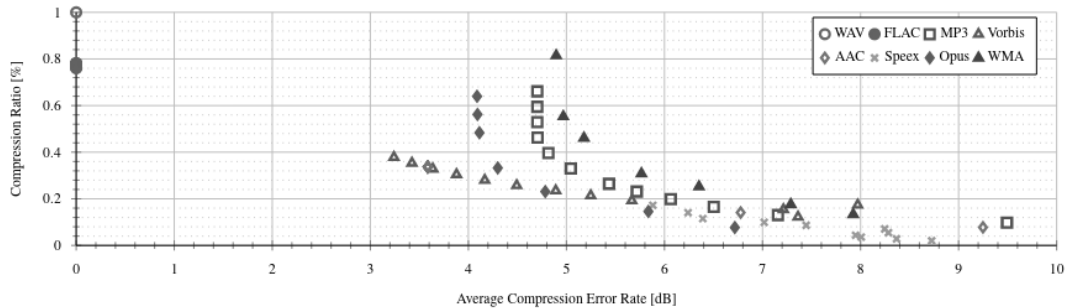
## Compresstion error



**Figure 3** - Average compression error rate for each codec and bit-rate. The bit-rate is increasing from left to right, see Table 2.

Figure 50: I Siegert, AF Lotz, LL Duong, A Wendemuth, Measuring the impact of audio compression on the spectral quality of speech data, 2016

# Compression ratio vs compression error



**Figure 4** - Average compression ratio over average compression error rate for each codec and bit-rate.

Figure 51: I Siegert, AF Lotz, LL Duong, A Wendemuth, Measuring the impact of audio compression on the spectral quality of speech data, 2016

## Conclusion

“we recommend to use FLAC for all cases where the accuracy matters. In cases where a slight error is acceptable, we recommend Vorbis at 500 kbit/s” (Siegert et al 2016)

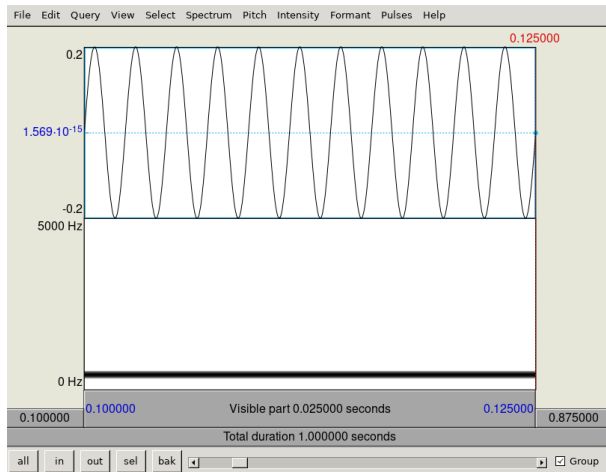


Figure 52: 440Hz sin wave



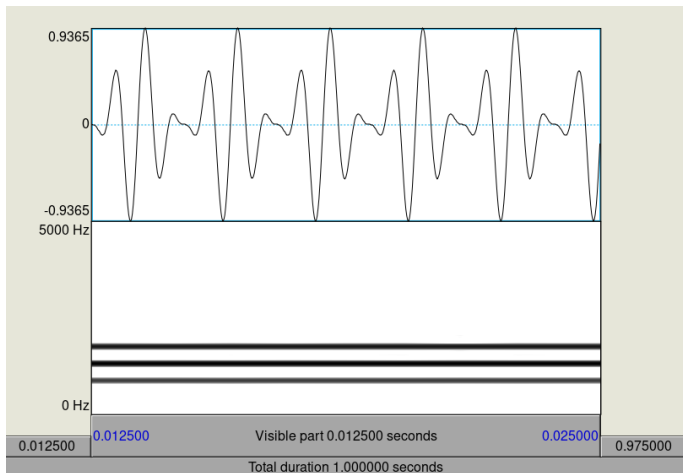
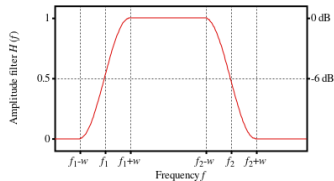


Figure 53:  $\frac{1}{4} \sin(2 \pi 880x) + \frac{1}{2} \sin(2 \pi 1320x) + \frac{1}{4} \sin(2 \pi 1760x)$

## Spectrum: Filter (pass Hann band)...

A command to modify every selected **Spectrum** object.

The complex values in the **Spectrum** are multiplied by real-valued sine shapes and straight lines, according to the following figure:



### Settings

**From frequency (Hz)** (standard value: 500 Hz)

the lower edge of the pass band ( $f_1$  in the figure). The value zero is special: the filter then acts as a low-pass filter.

**To frequency (Hz)** (standard value: 1000 Hz)

the upper edge of the pass band ( $f_2$  in the figure). The value zero is special: the filter then acts as a high-pass filter.

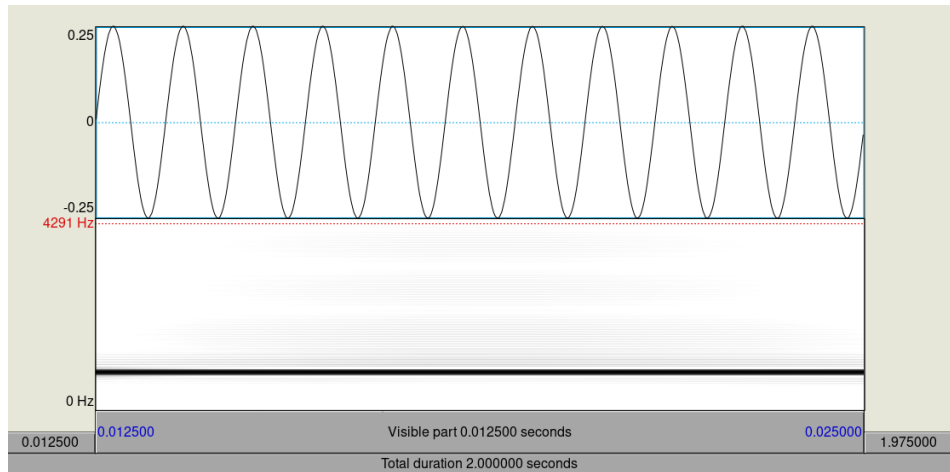
**Smoothing (Hz)** (standard value: 100 Hz)

the width of the region between pass and stop ( $w$  in the figure).

### Usage

Because of its symmetric Hann-like shape, the filter is especially useful for decomposing the Spectrum into consecutive bands. For instance, we can decompose the spectrum into the bands 0-500 Hz, 500-1000 Hz, 1000-2000 Hz, and 2000-“0” Hz:

Figure 54: Filter pass Hann band

Figure 55:  $f_1=780$ ,  $f_2=980$ ,  $w=100$

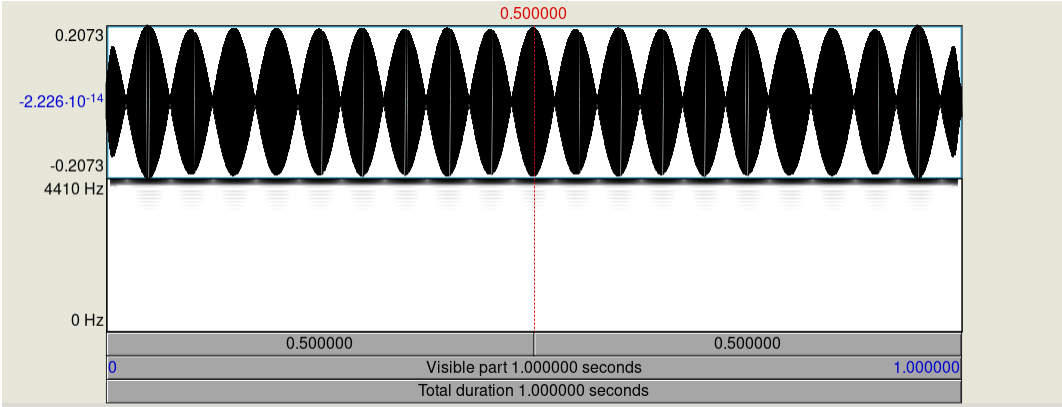


Figure 56: 4400Hz sin wave

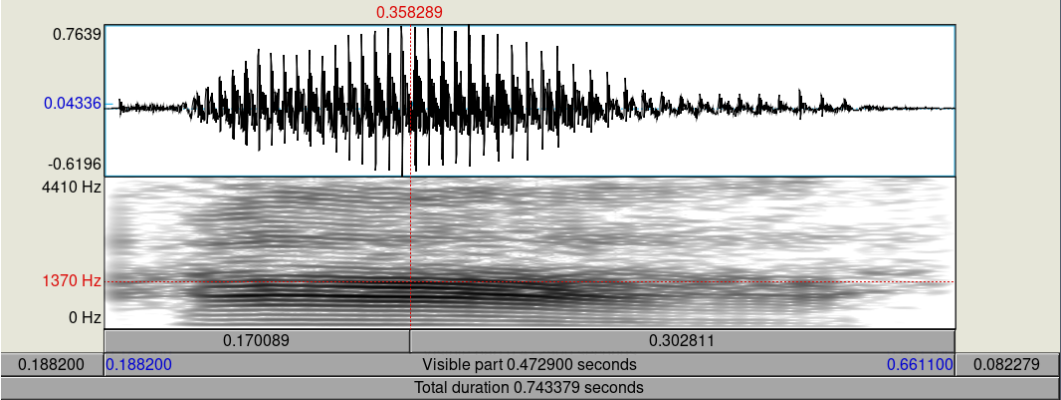


Figure 57: ah

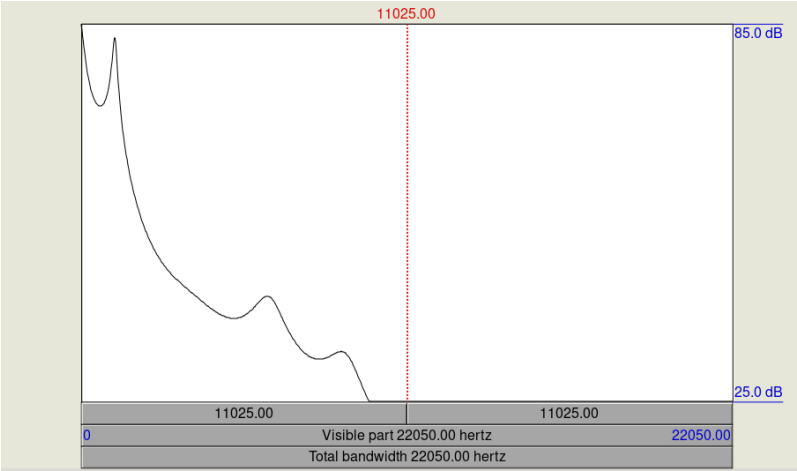


Figure 58: ah

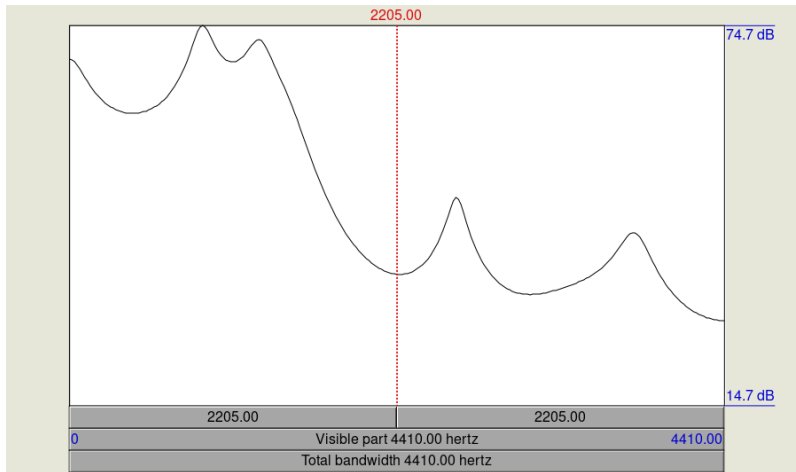


Figure 59: ah

THE END