

# Physics and Neurophysiology of Hearing

H.G. Dosch, Inst. Theor. Phys. Heidelberg

- I Signal and Percept
- II The Physics of the Ear
- III From the Ear to the Cortex
- IV Electrophysiology

## Part I: Signal and Percept

1) The time scales in hearing

2) Relation between Physics and Sensation

(Psychophysics)

a) Propagation and Production of Sound

b) The principal relations and caveats

c) A Byte of Signal processing (mainly Fourier etc)

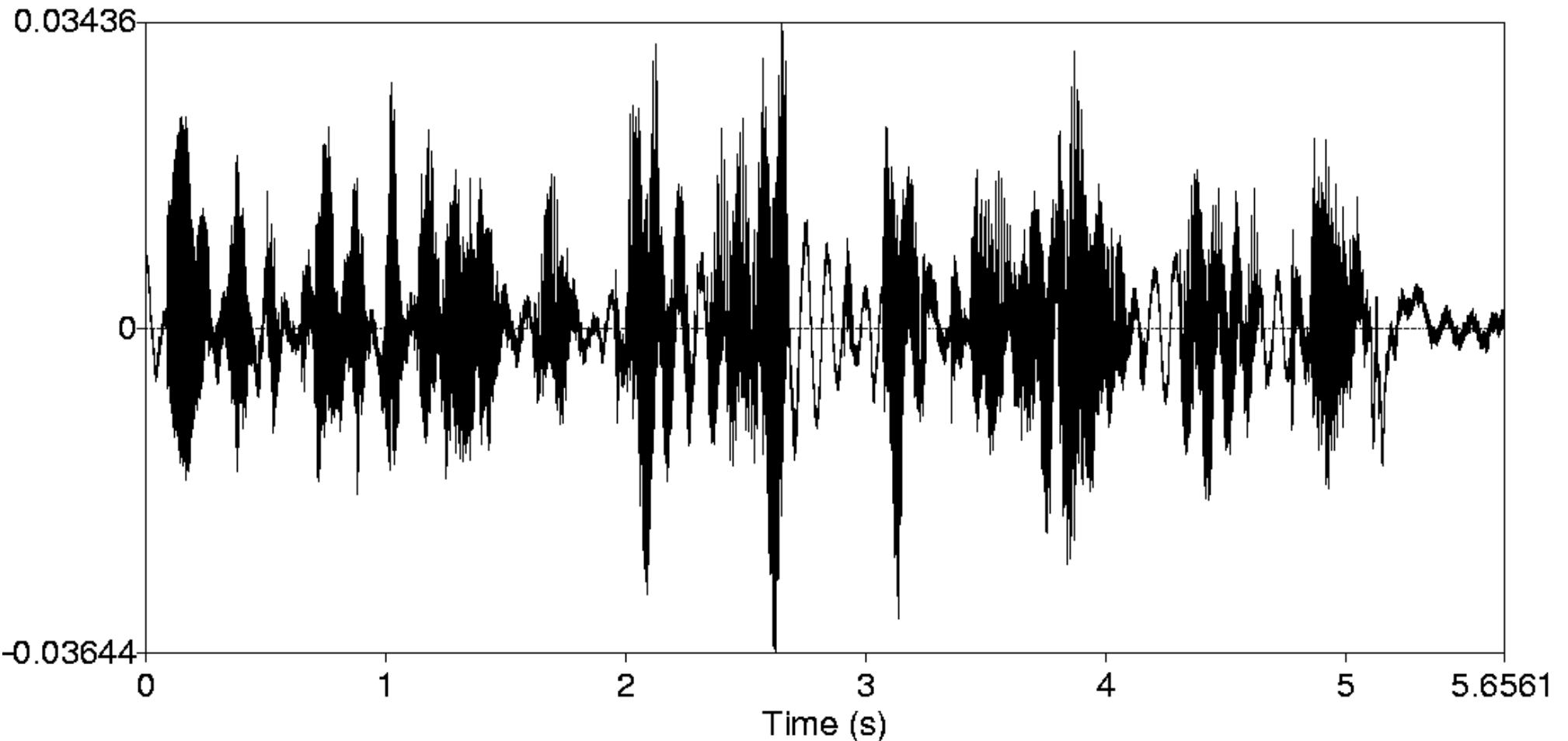
d) Application to acoustics: Ohms Law

e) Theory of Musical Consonance

f) Fusion of Harmonics and Fundamental Tracking

Sound pressure curve of the sentence:

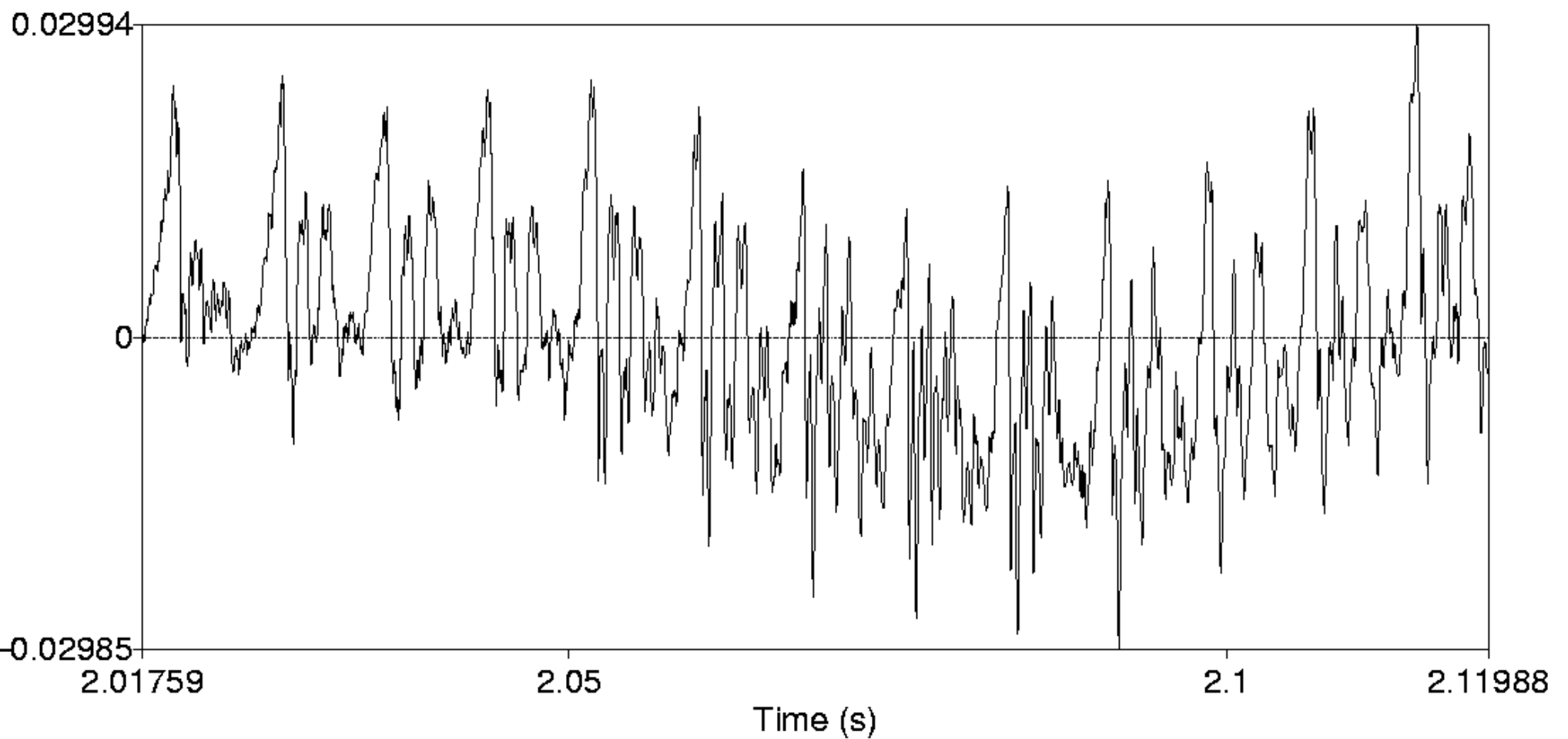
**each acoustical signal can be mapped unto a sound pressure curve**



Typical time scale of variation 0.1 s and larger

sound-pressure-example.wav in praat1.collection

selection of the previous sentence: mapped



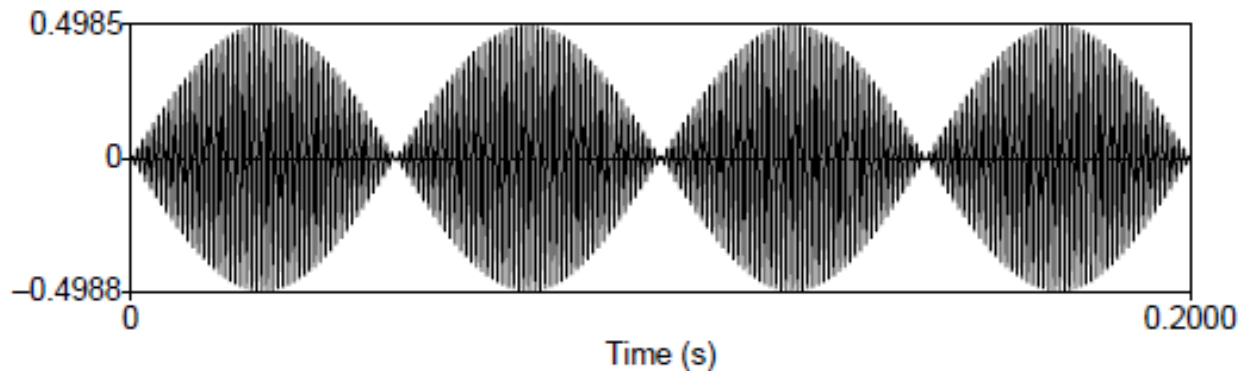
Typical time scale of variation: 10 ms

sound-pressure-selection.wav in praat1.collection

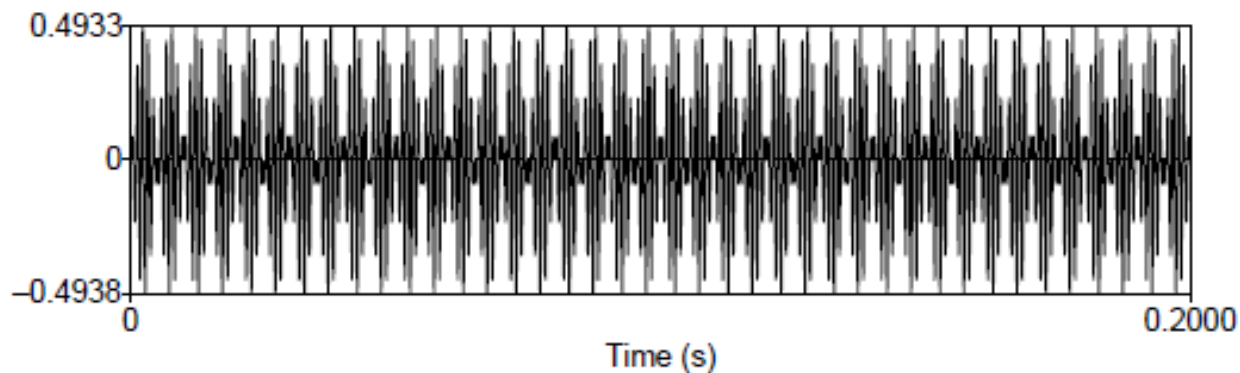
Amplitude-modulated sinusoidal tone with carrier frequency  $\nu_0$  and modulation frequency  $\nu_m$ .

$$p(t) = \sin(2\pi\nu_m t) \sin(2\pi\nu_0 t)$$

$$\nu_0 = 1000 \text{ Hz}$$

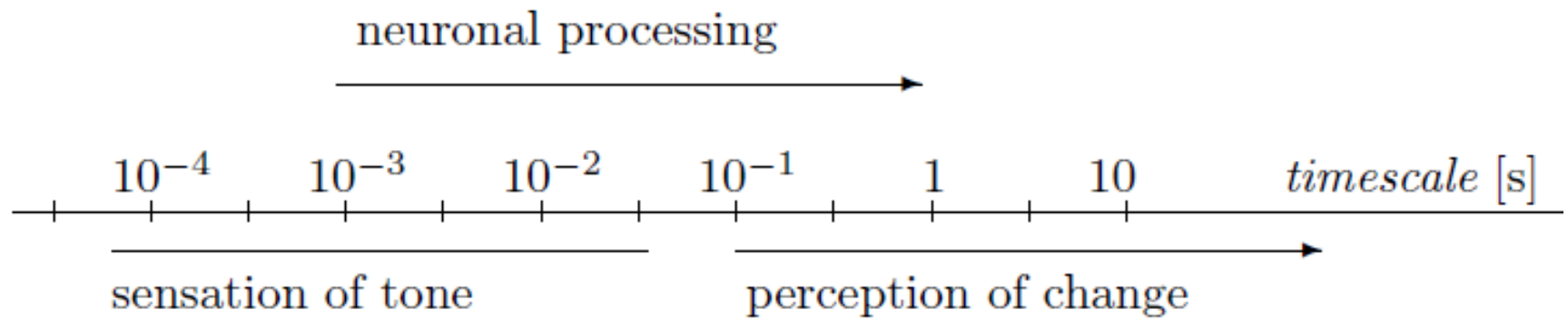


$$\nu_m = 10 \text{ Hz}$$

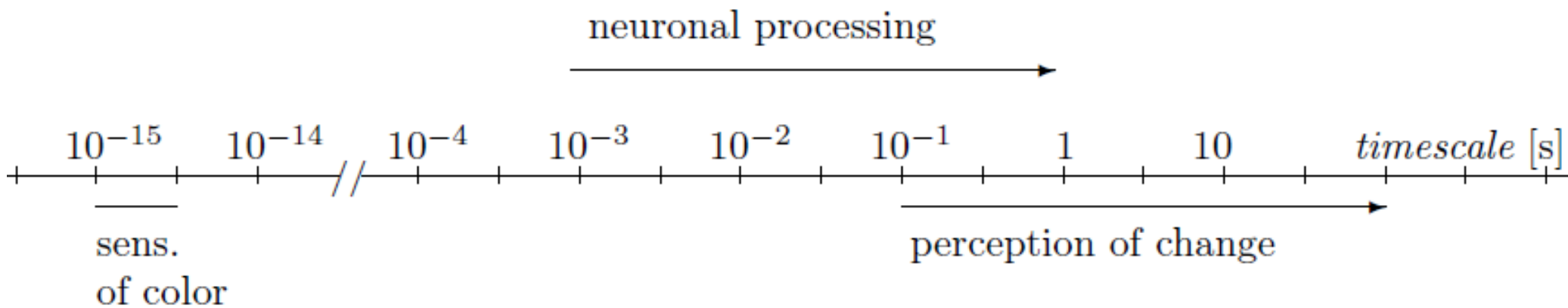


$$\nu_m = 100 \text{ Hz}$$

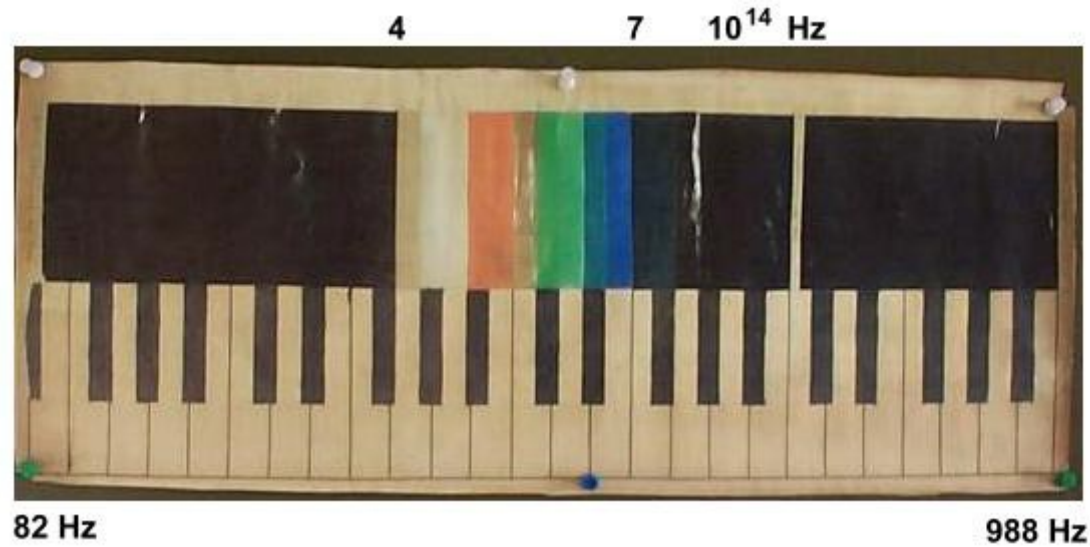
101praat:  
sinfreqmod10.wav  
sinfreqmod100.wav



Time scales in auditory processes



Time scales in visual processes

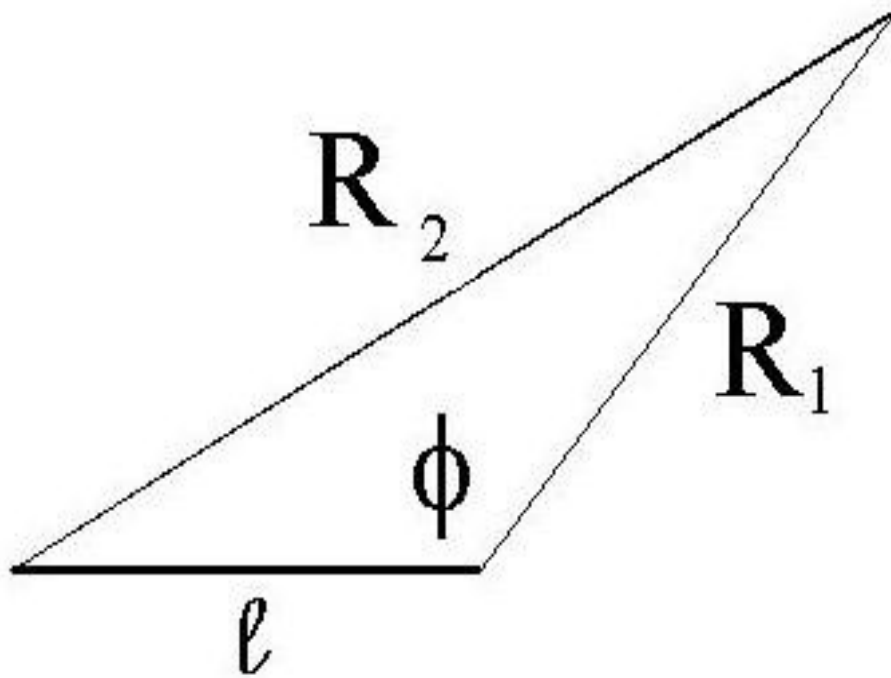


Original collage by Helmholtz to illustrate the optical and acoustical spectrum.

The red has faded away

The reception of sound waves in the ear is achieved by mechanical means on the basilar membrane, the reception of light by chemical processes in the retina.

In the following only concerned on the time variations in the ms range.



$$R_2^2 = l^2 + R_1^2 - 2lR_1 \cos \phi$$

$$d = R_2 - R_1 \approx l \cos \phi$$

$$\delta T = \frac{\delta d}{c_L} \approx -\frac{l}{c} \sin \phi \delta \phi$$

time resolution for angular resolution of 10 degrees

$$\frac{l}{c} \sin \phi \delta \phi \approx \frac{0.2}{330} \cdot \frac{2\pi}{360} \cdot 10 \approx 10^{-4} \text{ s}$$

shorter than tact of neurons



# 2

## Relation between Physics and Sensation Signal-Percept

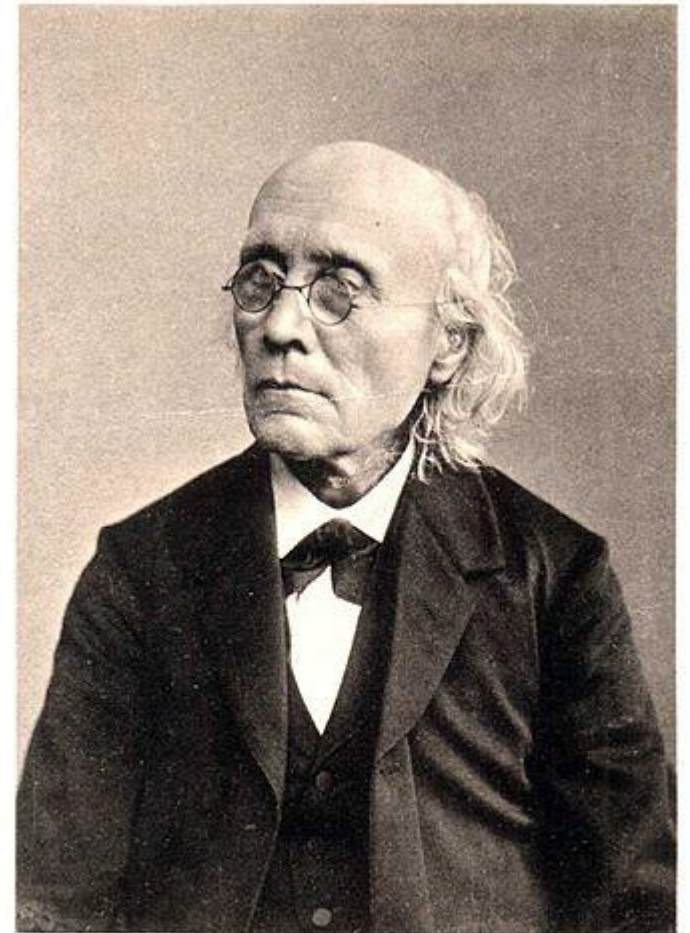
Psychophysics

Principal perceived properties of sound:

Loudness (volume)

Pitch

Timbre



*G. F. Fechner*

# 2a

$p(x)$ $v(x)$	$p(x) + dp$ $v(x) + dv$	$m = \rho F dx$

Propagation and generation of sound

Newton's Law  
in our case

$$m\dot{v} = K$$

Conservation of matter

$$\rho \dot{v} = -\partial_x p$$

Compressibility (adiabatic)

$$\partial_t \rho = -\partial_x(\rho v)$$

$$\frac{d\rho}{dp} = \frac{1}{c^2}$$

$c$  turns out to be the speed of sound in the material.

For sound:

1) the variations of density and and pressure are small compared to the prevailing values:

$$\Delta p \ll p_0, \Delta \rho \ll \rho_0 \text{ (e.g. atmospheric conditions).}$$

$$2) \dot{v} = \partial_t v + \partial_x v v \approx \partial_t v$$

Then we obtain:

$$\partial_t \rho = \frac{d\rho}{dp} \partial_t p = -\rho_0 \partial_x v$$

$$\frac{1}{c^2} \partial_t^2 p = -\rho_0 \partial_t \partial_x v = \partial_x^2 p$$

$$\frac{1}{c^2} \partial_t^2 p - \partial_x^2 p = 0$$

(Wave equation in 1+1 dimensions)

General solution (D'Alembert, ca 1750) :

$$p(x, t) = p_+(x + xct) + p_-(x - ct)$$

Also valid for strings, but there  $c$  related to the properties of the string (tension, diameter etc).

Solution with harmonic time dependence,  $\sim e^{i2\pi\nu t}$ , **without** boundary conditions

$$p(x, t) = Ae^{-2\pi i\nu t + ikx + i\delta}$$

$\nu$  Frequency,  $k = 2\pi/\lambda = 2\pi\nu/c$  wave number ( $\lambda$  wave length).

**With** boundary conditions, e.g.  $\rho(0, t) = \rho(L, t) = 0$   
special solution

$$p(x, t) = \sin\left(\frac{n\pi}{L}x\right) \cos\left(\frac{n\pi c}{L}t\right)$$

organ pipe with two open ends or vibrating string

relation between variation of  $p$  and  $v$  for the wave solution:

$$\omega\rho_0v = -kp \text{ or } p = -Zv$$

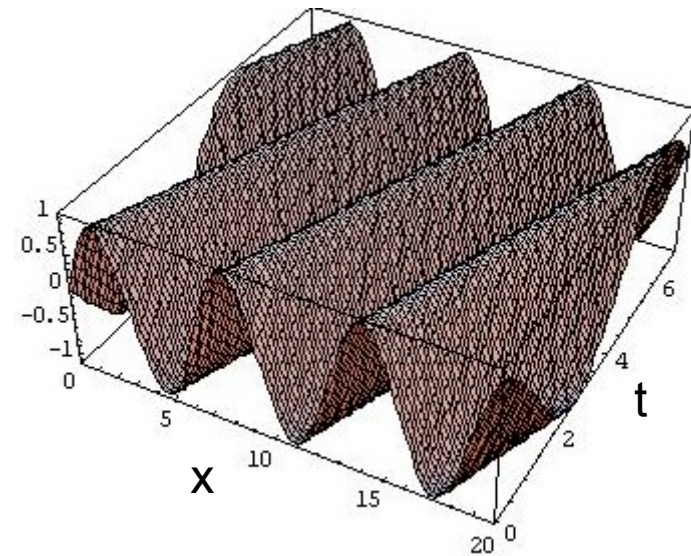
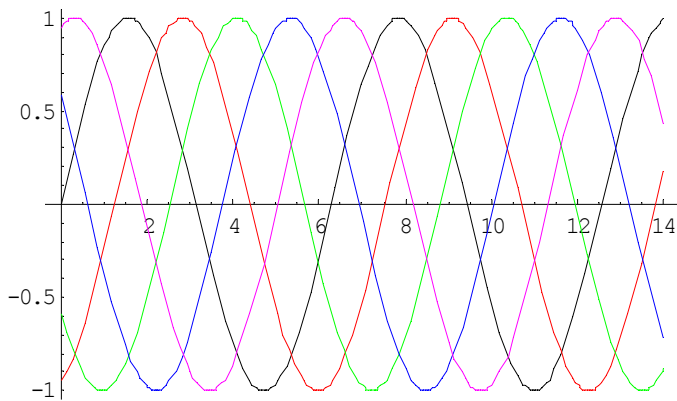
$Z = \rho_0c$  is called the impedance (wave resistance) of the medium

(cf with  $U = IR$ ).

Special solution, **without** boundary conditions (propagation of sound in space):

$$\rho(x, t) = A/c^2 \sin(2\pi\nu t - kx)$$

$\nu$  Frequency,  $k = 2\pi/\lambda = 2\pi\nu/c$  wave number ( $\lambda$  wave length). Wave propagating to the right.

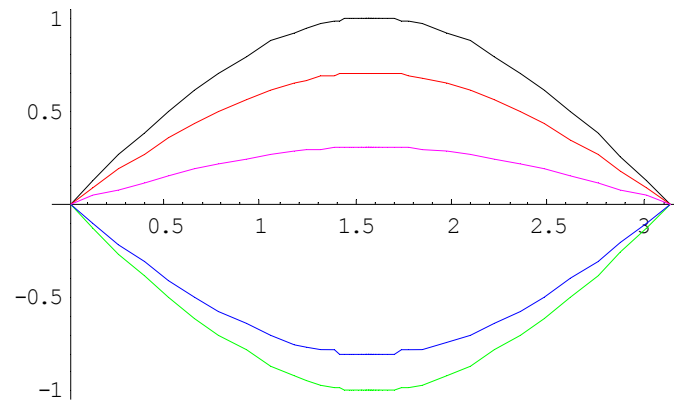


$$\partial_t p = c^2 \rho_0 \partial_x v = -c^2 \rho_0 \frac{k}{\nu} \partial_t v = -Z \partial_t v$$

$Z = \rho_0 c$  is called the impedance (wave resistance) of the medium (cf with  $U = IR$ ).

**With** boundary conditions, e.g.  $\rho(0, t) = \rho(L, t) = 0$   
special solution

$$\rho(x, t) = \sin\left(\frac{n\pi}{L} x\right) \cos\left(\frac{n\pi c}{L} t\right)$$



organ pipe with two open ends or vibrating string

## 2b Relation between acoustical signal (sound pressure) and perception

(to be refined later)

Amplitude of sp

Loudness

If the change of the sensation,  $ds$ , is proportional to the relative change of the physical stimulus  $R$

$$ds = \frac{dR}{R} \quad \text{then } s = \log \frac{R}{R_0}$$

$R_0$  is the threshold of sensation.

Therefore in acoustics the measure of the energy of sound pressure level is the decibel (dB).

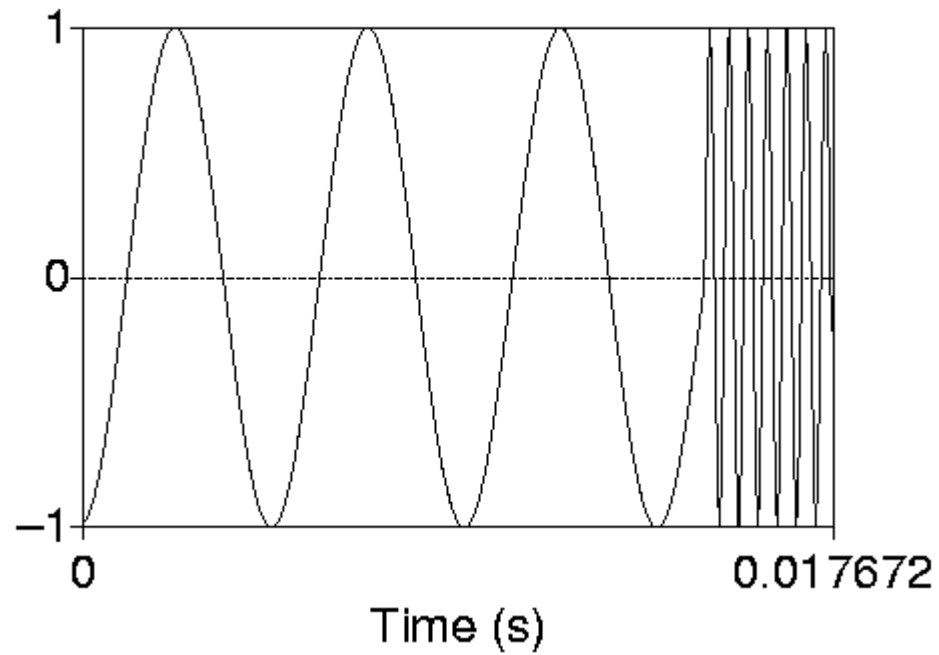
If  $R_0$  is amplitude (actually rms) of the sound pressure level at hearing threshold (at 2000 Hz) then dB (SPL) is given as

$$10 \log_{10} \left[ \frac{R^2}{R_0^2} \right] = 20 \log_{10} \left[ \frac{R}{R_0} \right]$$

Relation Energy-loudness  
not so simple

Periodicity

pitch of the tone

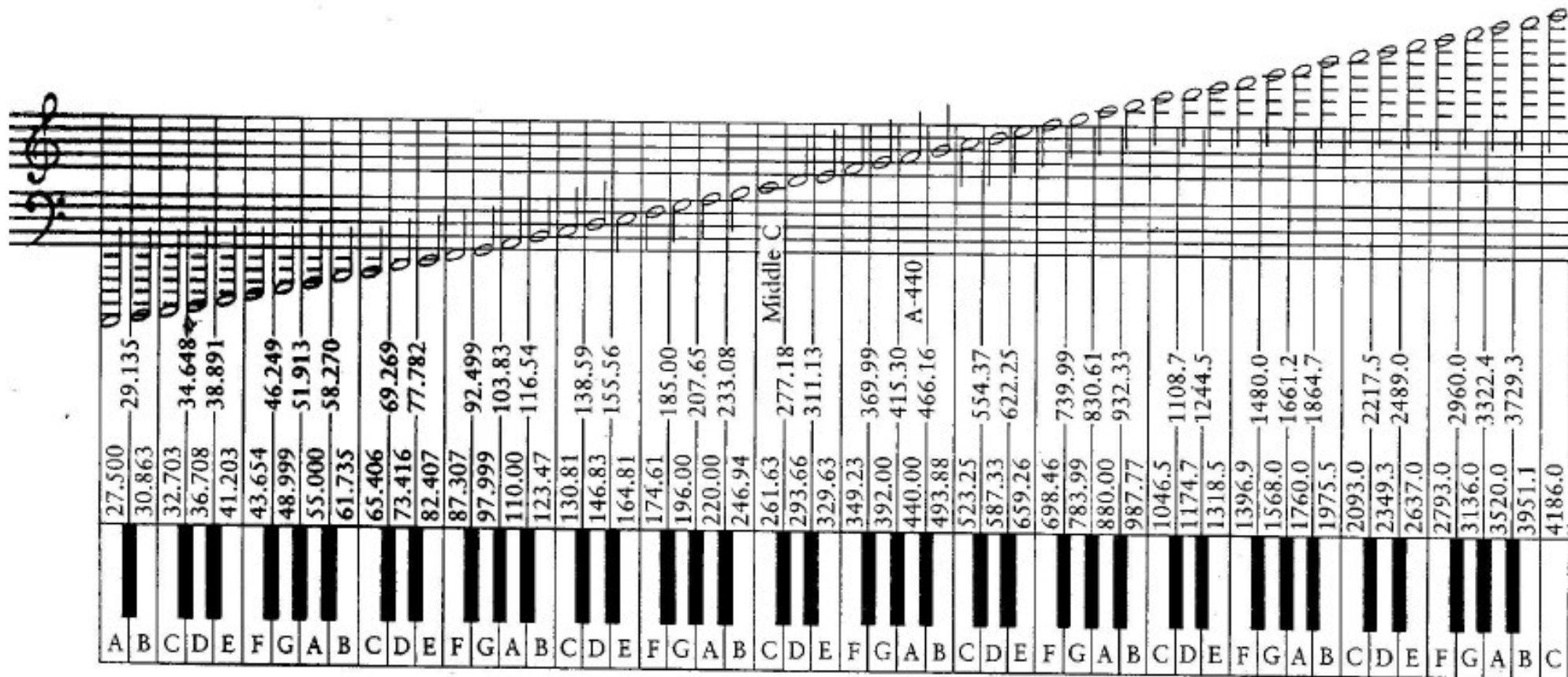


101praat: tief-hoch.wav

again: real life is a bit more complicated

101praat: sound-440\_3-5, 366\_4\_6





A2

A1

A

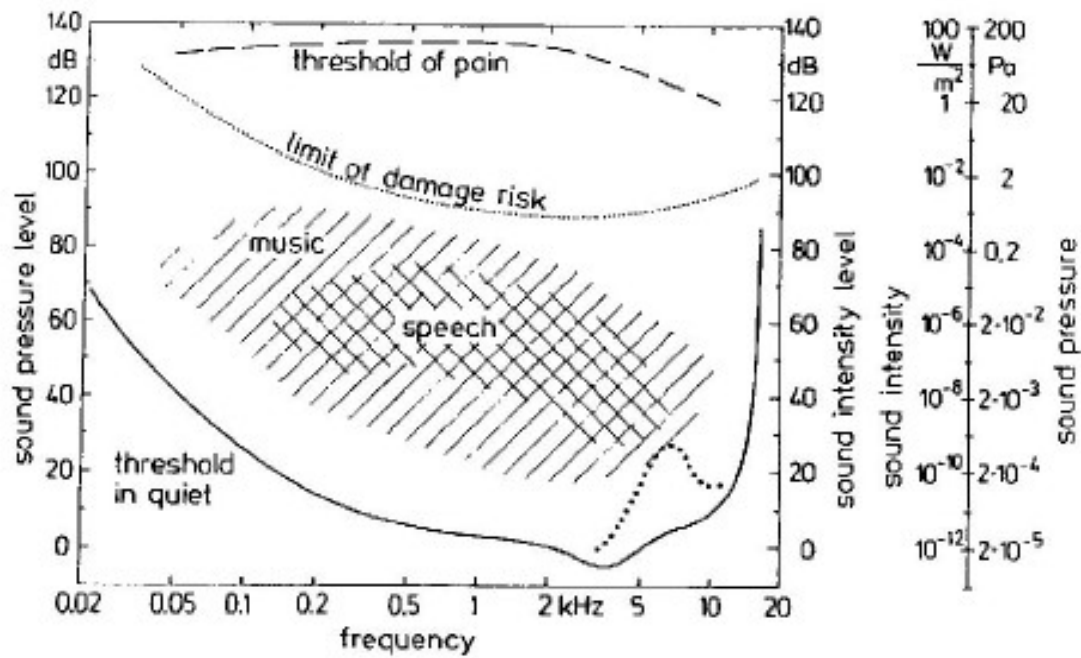
a

a'

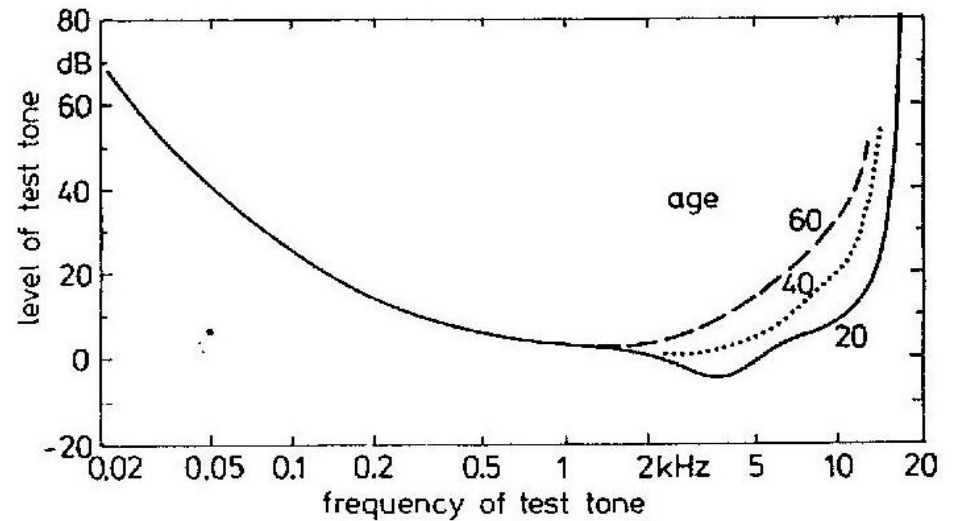
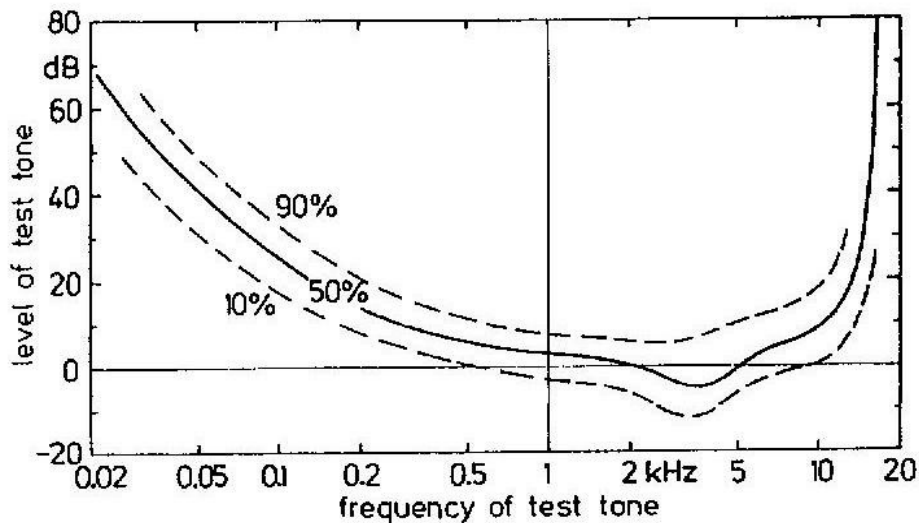
a''

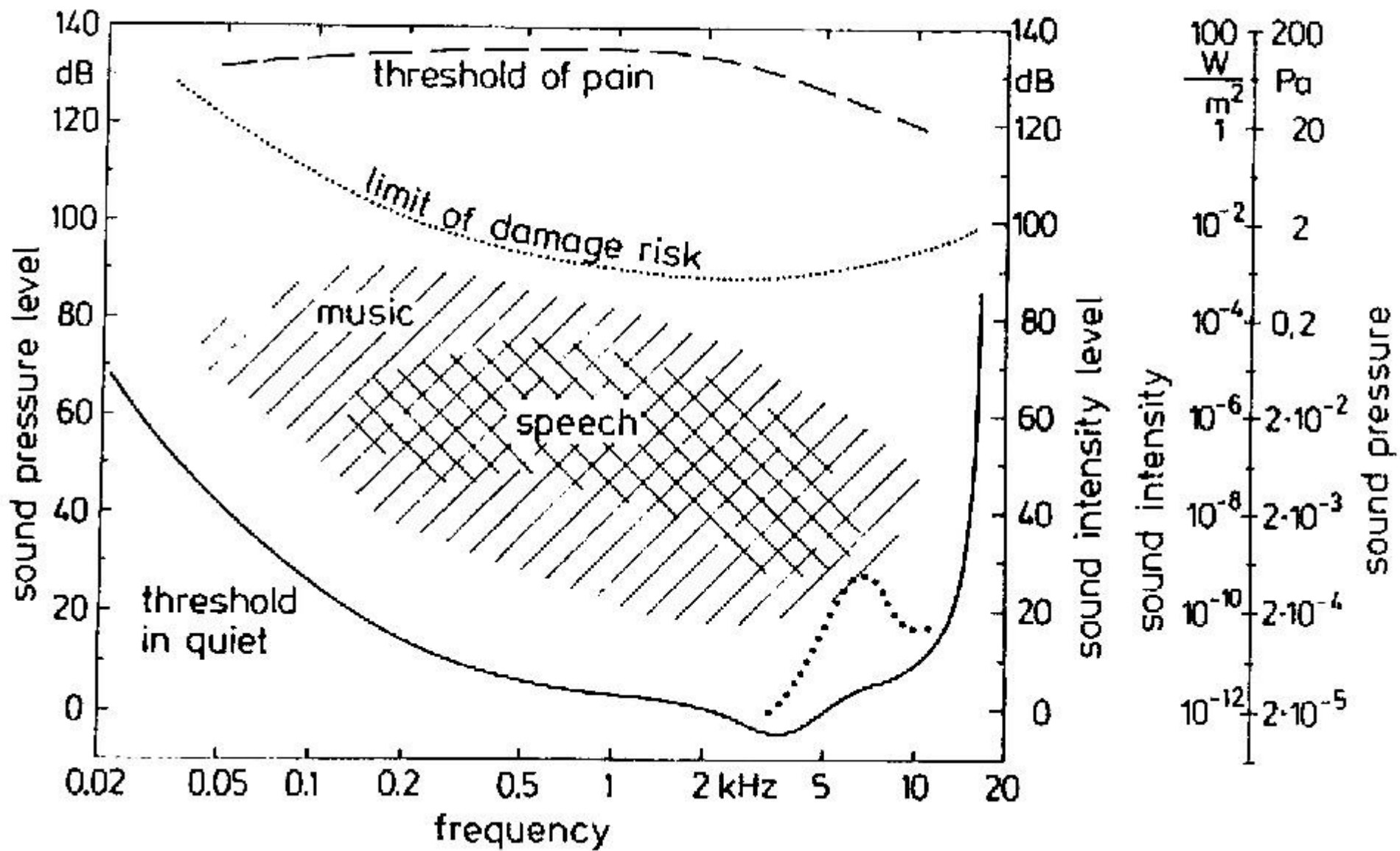
a'''

a''''



Intensität $W/cm^2$	Lautstärke
$10^{-4}$	Schmerzgrenze
$10^{-7}$	<i>fff</i>
$10^{-10}$	<i>mf</i>
$10^{-13}$	<i>ppp</i>
$10^{-16}$	Hörschwelle





Most complicated sensation: timbre

not (approximately) 1 dimensional like volume and pitch  
but many properties:

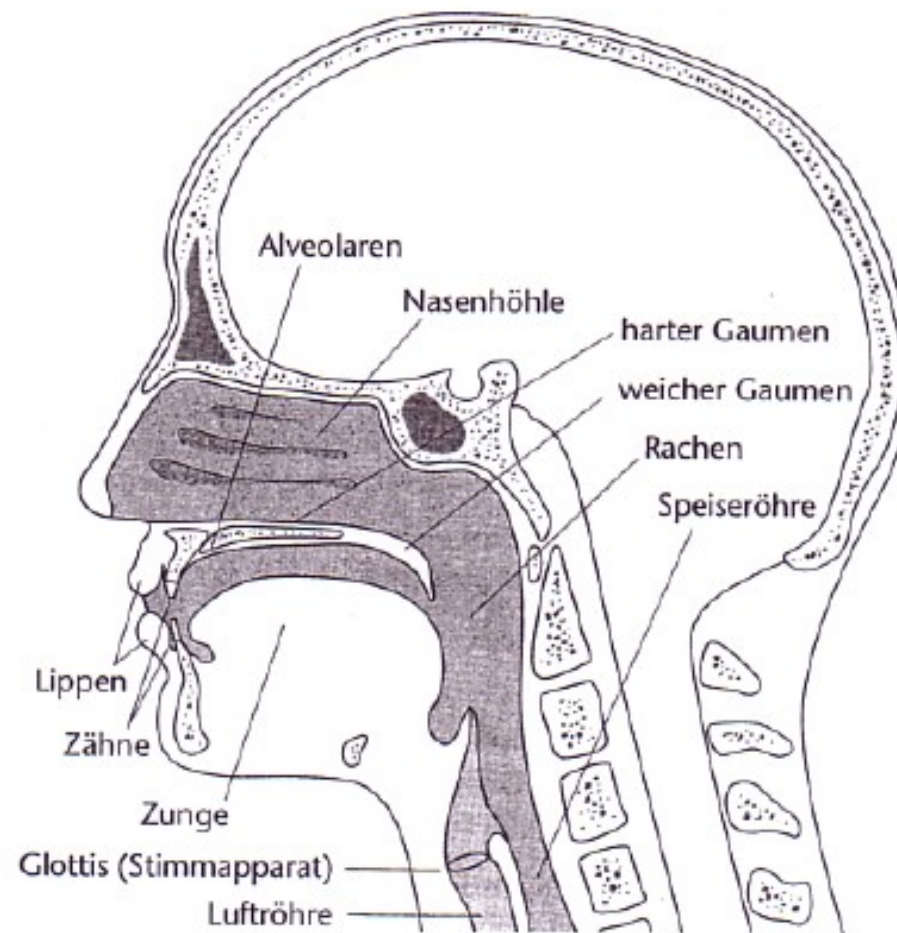
full, thin, soft, harsh .....

general relation:

form of the SP curve      timbre

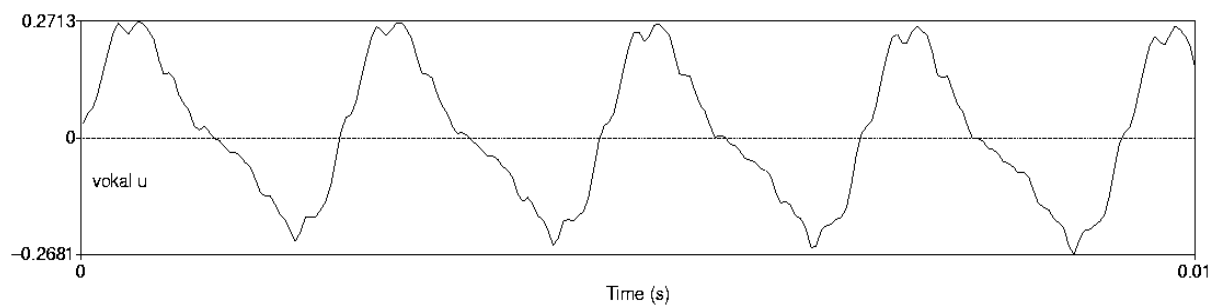
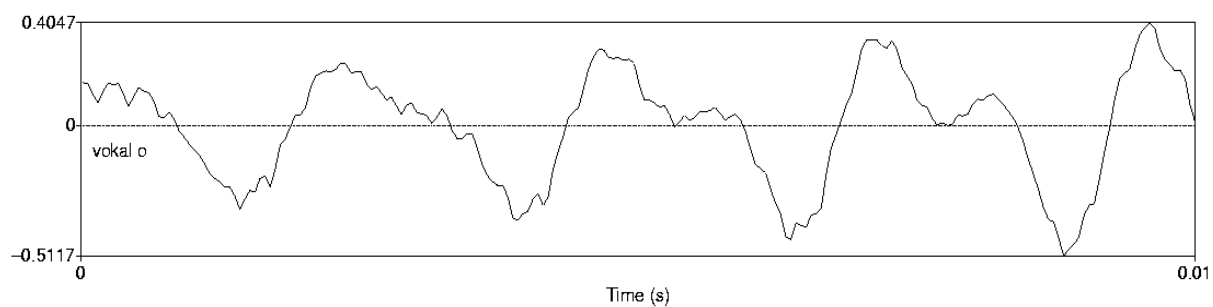
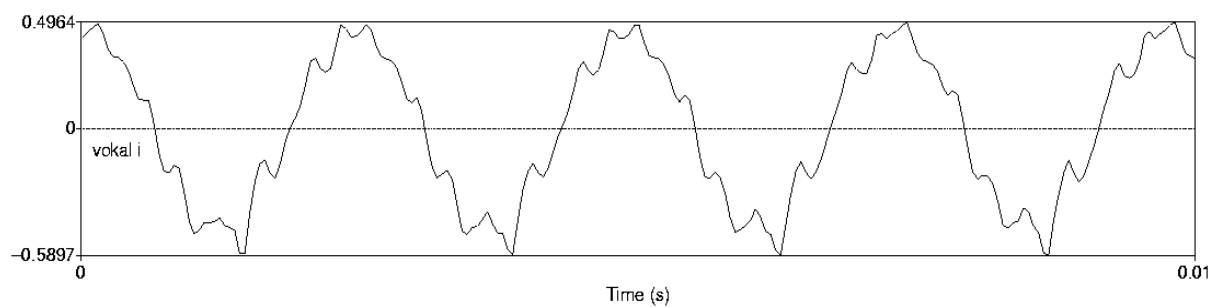
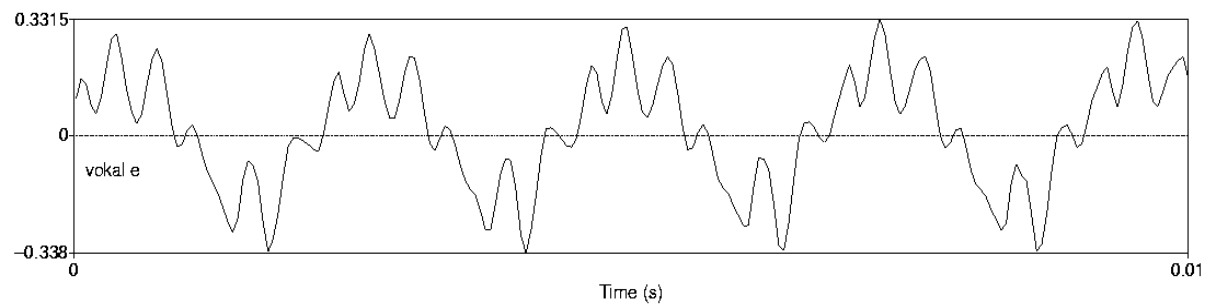
A very important distinction is the different timbre of different vowels

# The human vocal tract

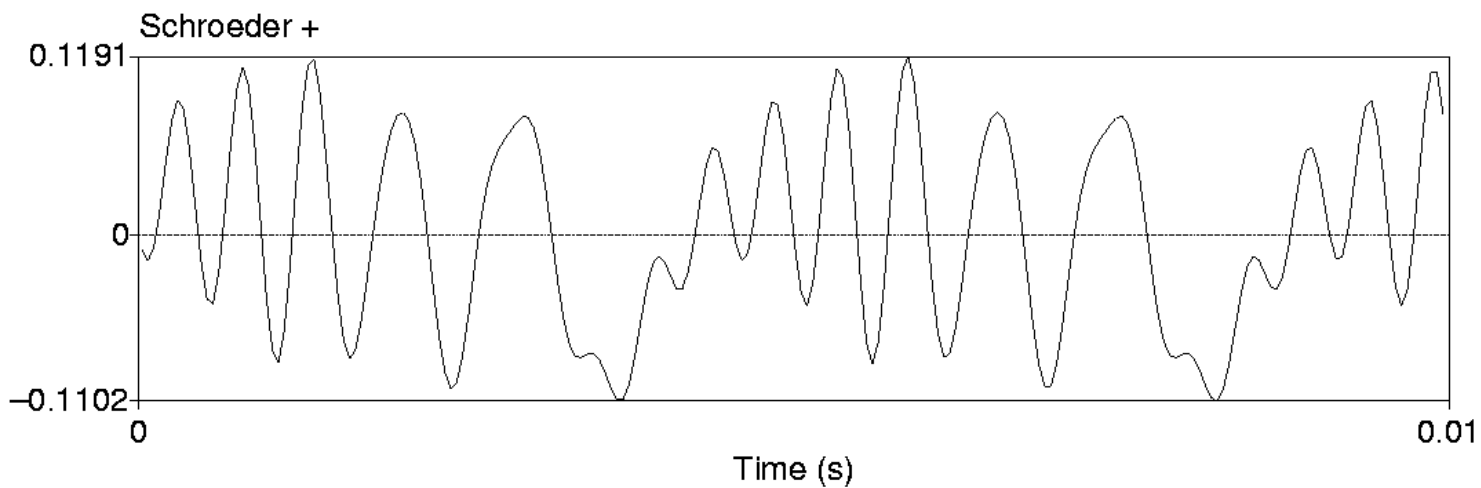


*Abb. 12.1 Zum Stimmtrakt gehören Nasen- und Mundhöhle und der Rachen sowie Bestandteile, die sich bewegen, etwa Zunge, Lippen und Stimmbänder.*

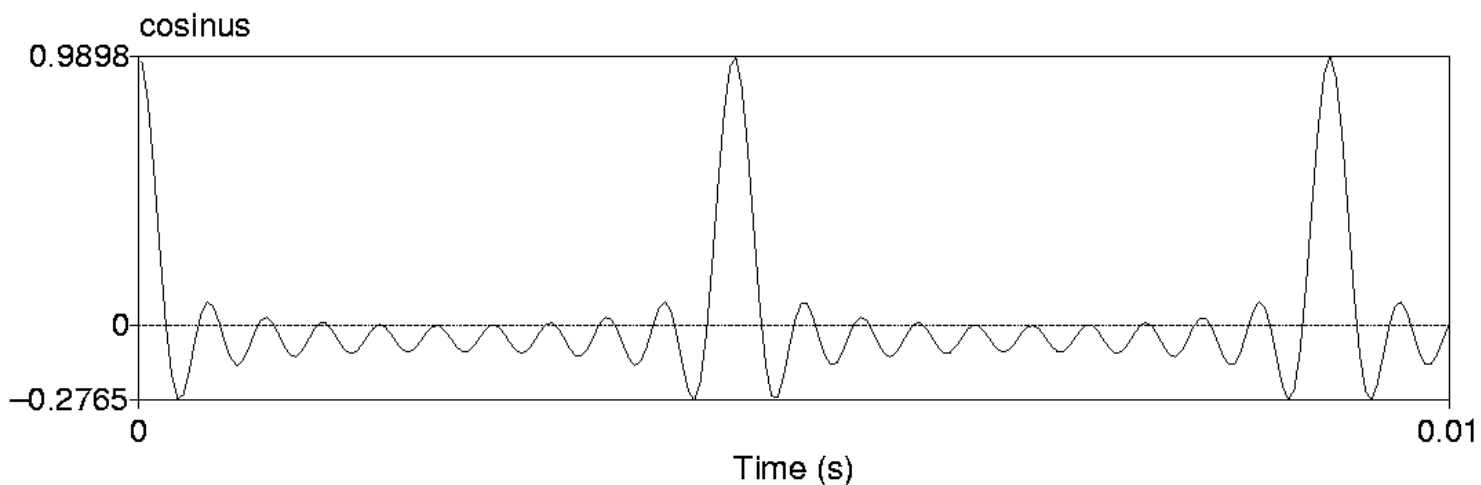
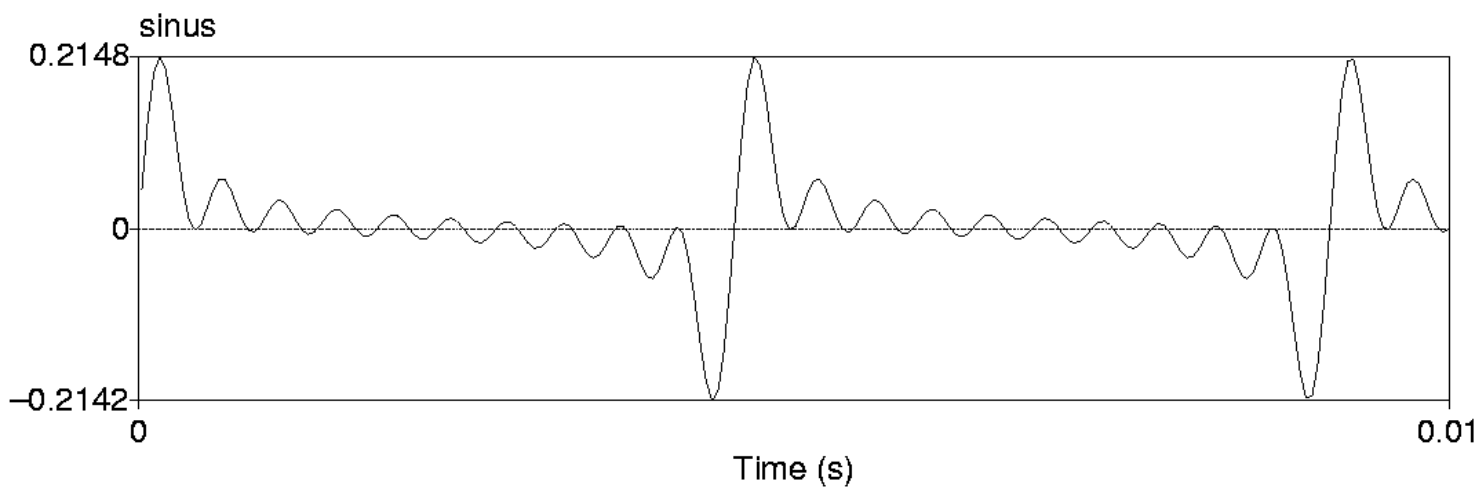
The vowels sound very different, though the sp curves look similar:



101praat: e\_reg - u\_reg



The sound pressure curves look different, but the sounds are practically indistinguishable



101prat:  
schroederplus10,  
sin10,cos10

# Fourier and wavelet transforms

Fourier transform:

$$\tilde{f}(\omega) = \int_{-\infty}^{\infty} dt f(t) e^{i\omega t}$$

magic  
formula

$$\int \frac{d\omega}{2\pi} e^{i\omega\alpha} e^{-i\omega t} = \delta(t - \alpha)$$



$$f(t) = \int_{-\infty}^{\infty} \frac{d\omega}{2\pi} \tilde{f}(\omega) e^{-i\omega t}$$

The squared modulus

$$|\tilde{f}(\omega)|^2$$

is called *power spectrum*.




The Fourier transform is linear:  $\alpha f(t) + \beta g(t) \rightarrow \alpha \tilde{f}(\omega) + \beta \tilde{g}(\omega)$

If the function  $f(t)$  is real, i.e. if  $f(t) = f^*(t)$ , then  $\tilde{f}(\omega) = \tilde{f}^*(-\omega)$ .

Convolution:

$$(f * g)(\tau) = \int_{-\infty}^{\infty} dt dt' f(t)g(t')\delta(t + t' - \tau) = \int_{-\infty}^{\infty} dt f(t)g(\tau - t) = \int_{-\infty}^{\infty} dt f(\tau - t)g(t)$$


$$\int_{-\infty}^{\infty} dt f(t)g(t)e^{i\omega t} = \frac{1}{2\pi}(\tilde{f} * \tilde{g})(\omega)$$

# Signal Processing

$t$  time;  $\omega = 2\pi\nu$  circular frequency;  
 $\nu$  frequency

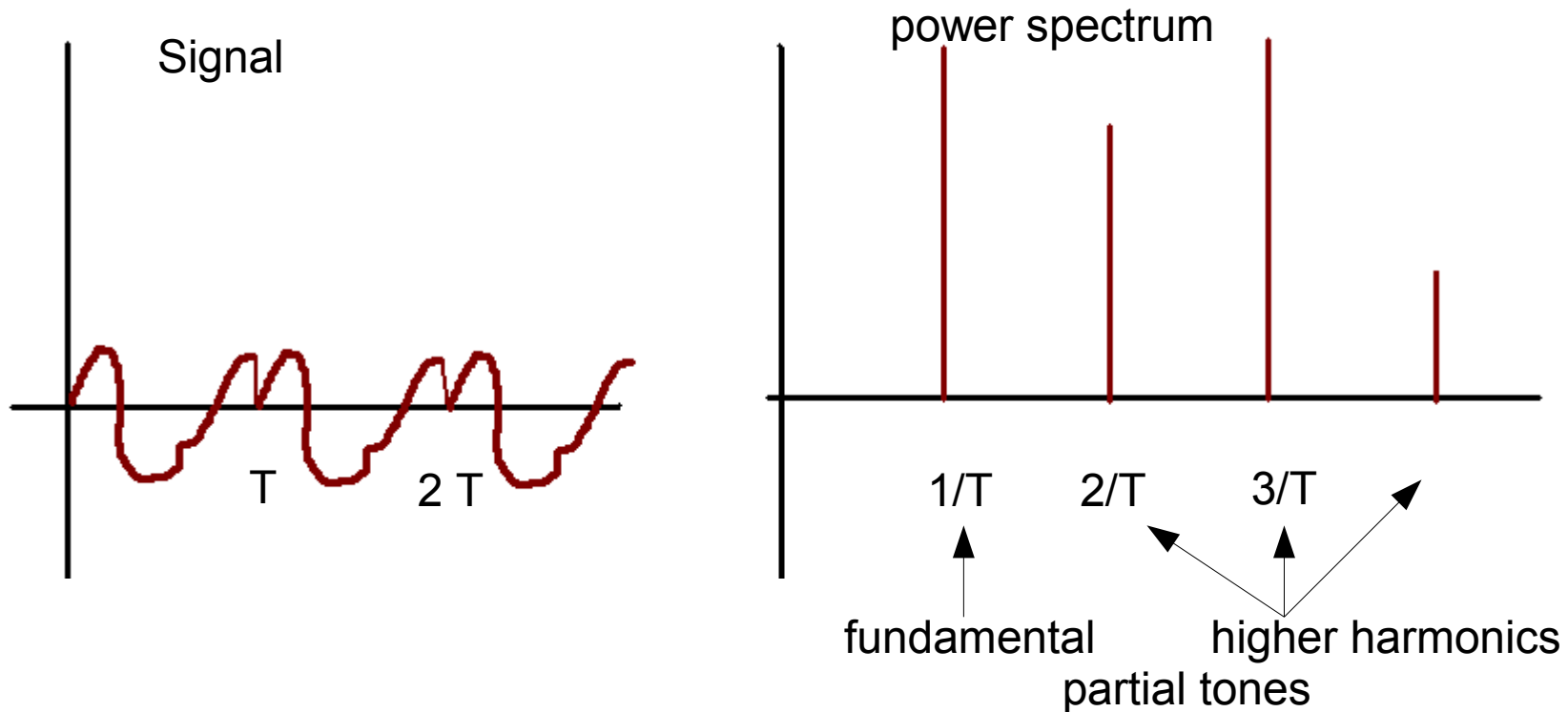
$p(t)$  Signal (e.g. sound pressure as function of time)

$\tilde{p}(\nu) = \int dt p(t) e^{i2\pi\nu t}$  spectral function

$|\tilde{p}(\nu)|^2$  (power) spectrum, normally displayed in logarithmic scale (dB) 10 db a factor 10 in power ( $\sqrt{10}$  in amplitude)

# Essential Theorem:

If the signal is periodic with period  $T$ , then the spectral function is different from zero only at  $n/T$ ,  $n$  integer. Harmonic spectrum



$$p(t) = p(t + T) \Rightarrow \tilde{p}(\nu) = \sum_n c_n \delta(\nu - \frac{n}{T})$$

Proof:

$$p(t) = \int dt \tilde{p}(2\pi\nu) e^{i2\pi\nu t} = \int dt \tilde{p}(2\pi\nu) e^{i2\pi\nu(t+T)}$$

$$\Rightarrow 2\pi\nu T = n$$

Only valid if  $\nu T = n$  (integer).

Fourier transform



signal

spectral function

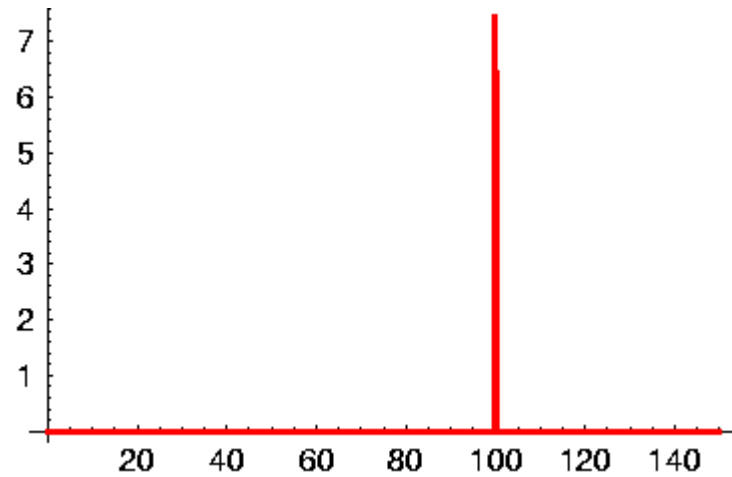
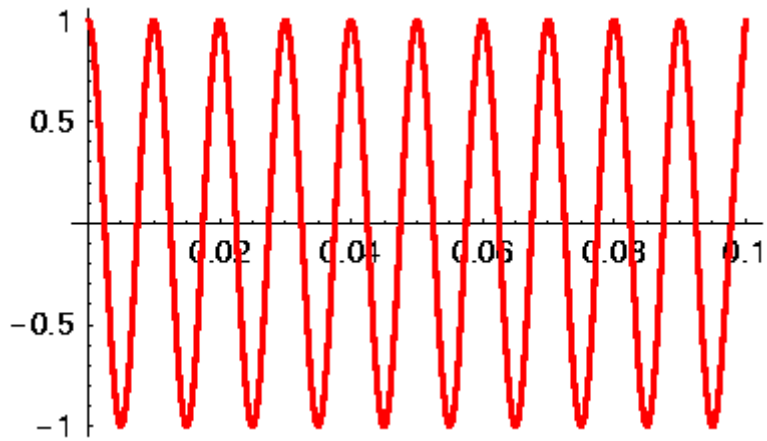
$$p(t) = \cos(2\pi\nu_0 t)$$

$$p(t) = \sin(2\pi\nu_0 t)$$

$$\tilde{p}(\nu) = \pi \delta(\nu - \nu_0) + \pi \delta(\nu + \nu_0)$$

$$\tilde{p}(\nu) = i\pi \delta(\nu - \nu_0) - i\pi \delta(\nu + \nu_0)$$

$$\nu_0 = 100 \text{ Hz}$$



Fourier transform



signal

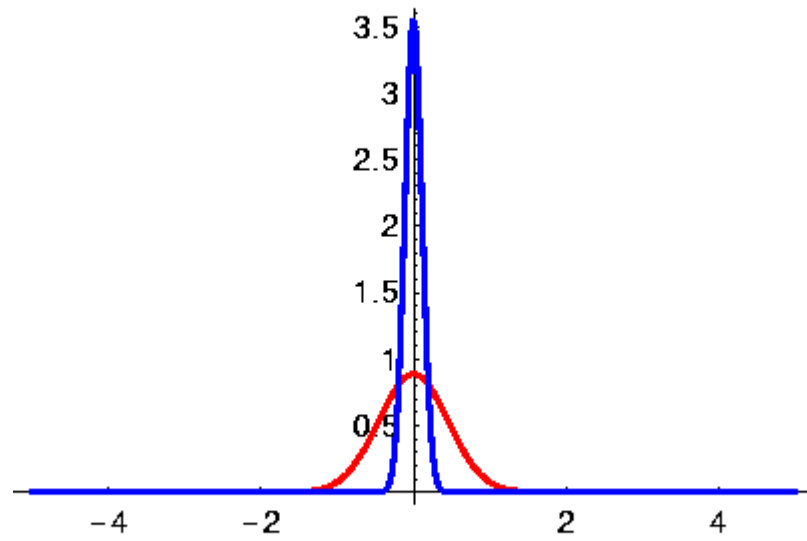
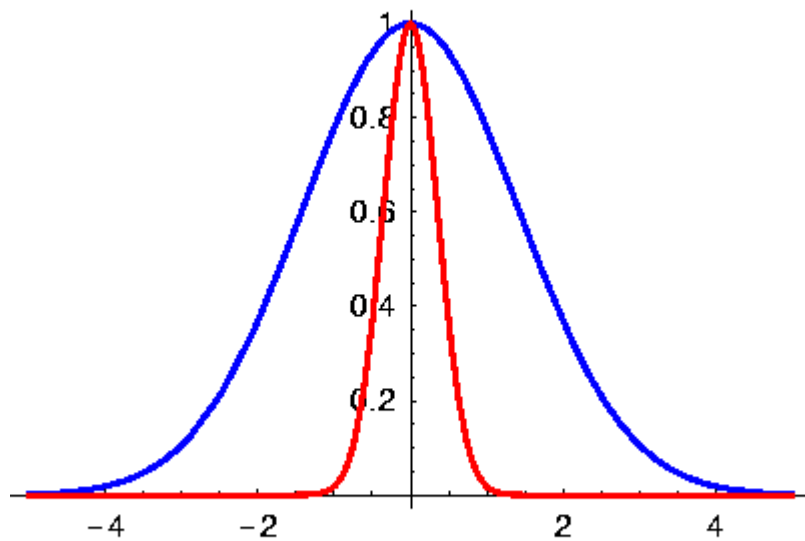
spectral function

$$p(t) = e^{-(t-c)^2/b^2}$$

$$\tilde{p}(2\pi\nu) = e^{i2\pi\nu c} b\sqrt{\pi} e^{-b^2(2\pi\nu)^2/4}$$

$b=0.5$

$b=2$



signal

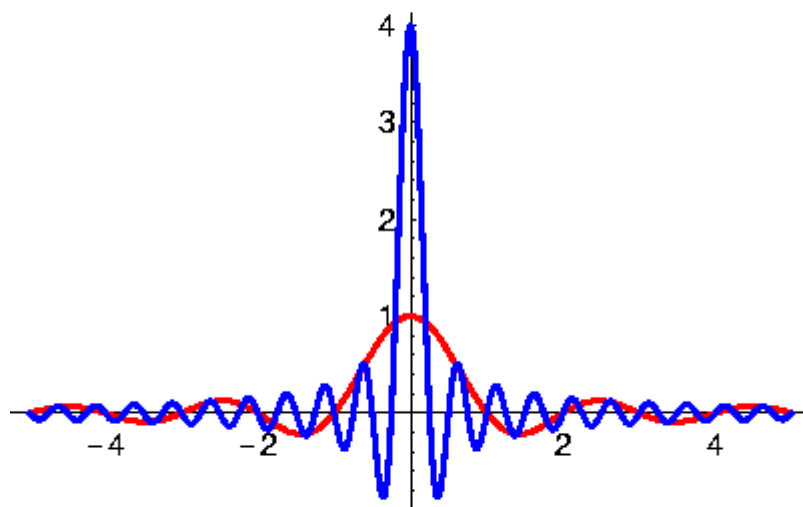
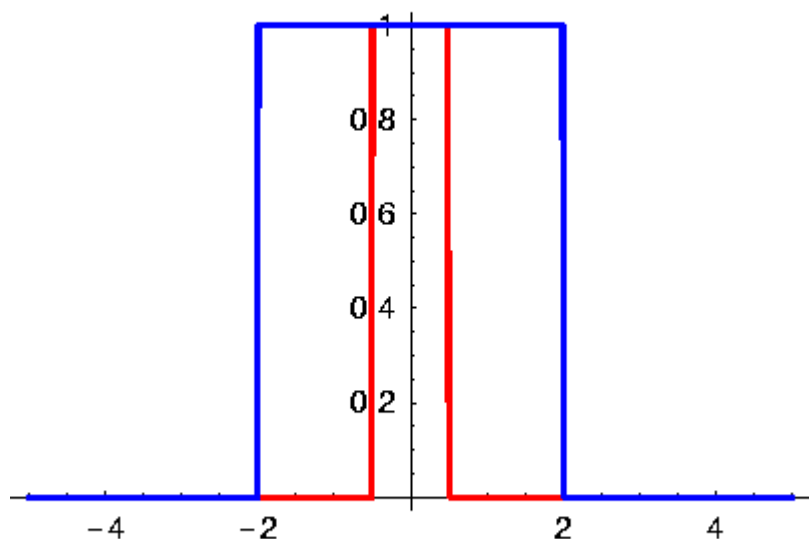
spectral function

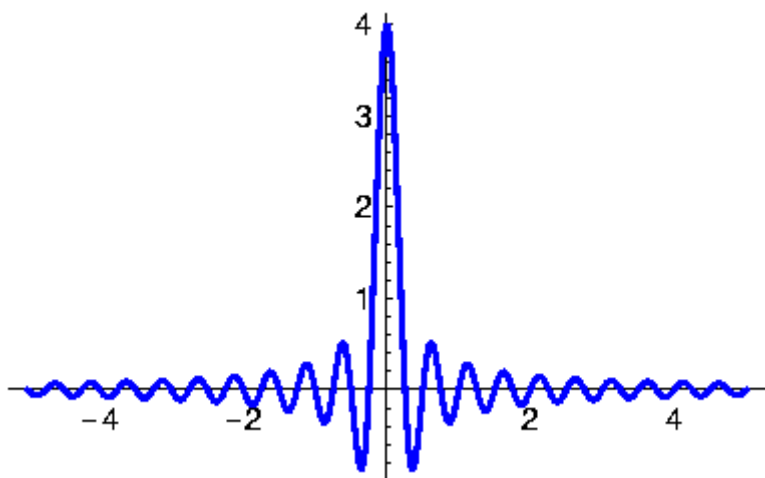
$$p(t) = \theta(t - c + b) \cdot$$

$$\theta(c + b - t)$$

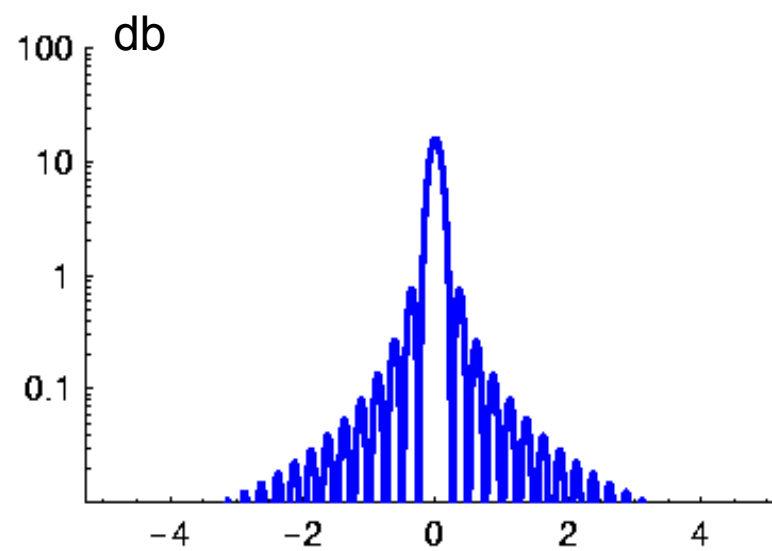
$$\tilde{p}(\nu) = e^{i2\pi\nu c} 2 \frac{\sin(b2\pi\nu)}{2\pi\nu}$$

c=0   b=0.5   b=2





spectral function



power spectrum



# Windowed Fourier Analysis, Wavelet analysis

Fourier analysis over a time window,  $g(t)$  :

$$\tilde{f}(t, \omega) := \int_{-\infty}^{\infty} dt' g(t - t') f(t') \exp[-i\omega t']$$

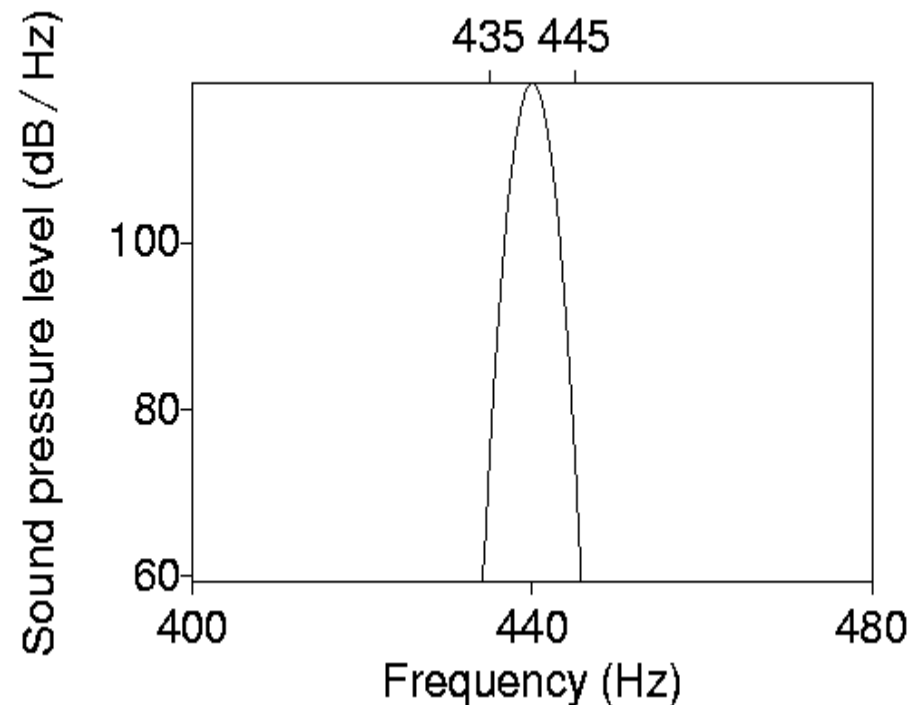
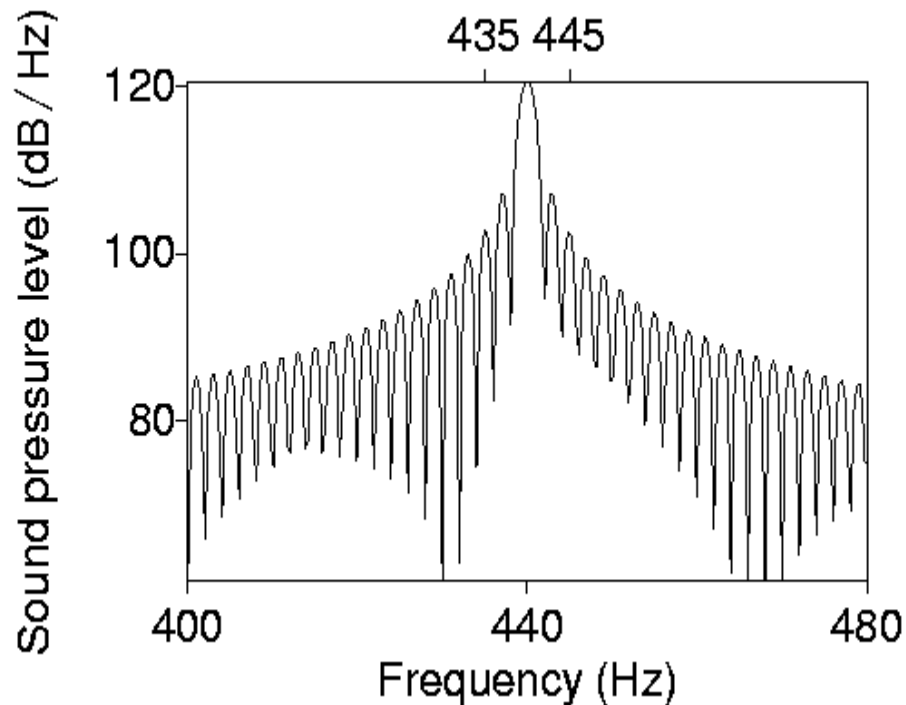
possible time windows    rectangular

$$g(t - t') = \theta(t - t')\theta(\Delta - (t - t')).$$

Gaussian window

$$g(t - t') = \frac{1}{\Delta\sqrt{\pi}} \exp\left[-\frac{(t - t')^2}{\Delta^2}\right].$$

Power spectrum of  $\sin(2\pi 440 t)$  for rectangular and Gaussian window ( $\Delta = 0.5s$ )



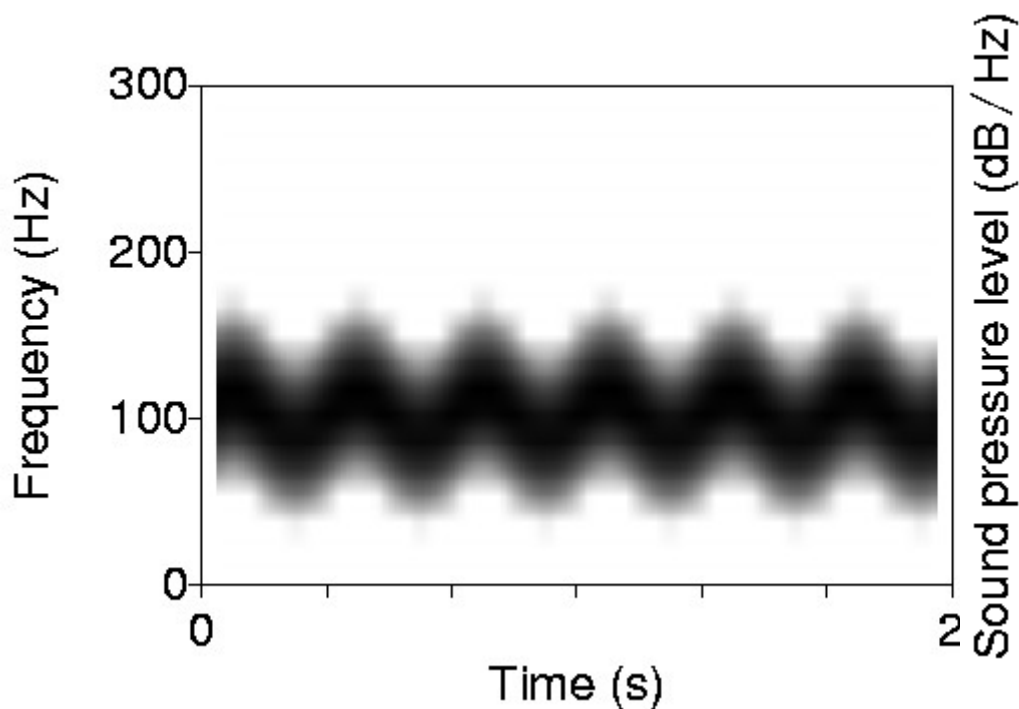
Representation windowed Fourier transform by Spectrogramm:

time as x-Axes, frequency as y axis and intensity as gray value

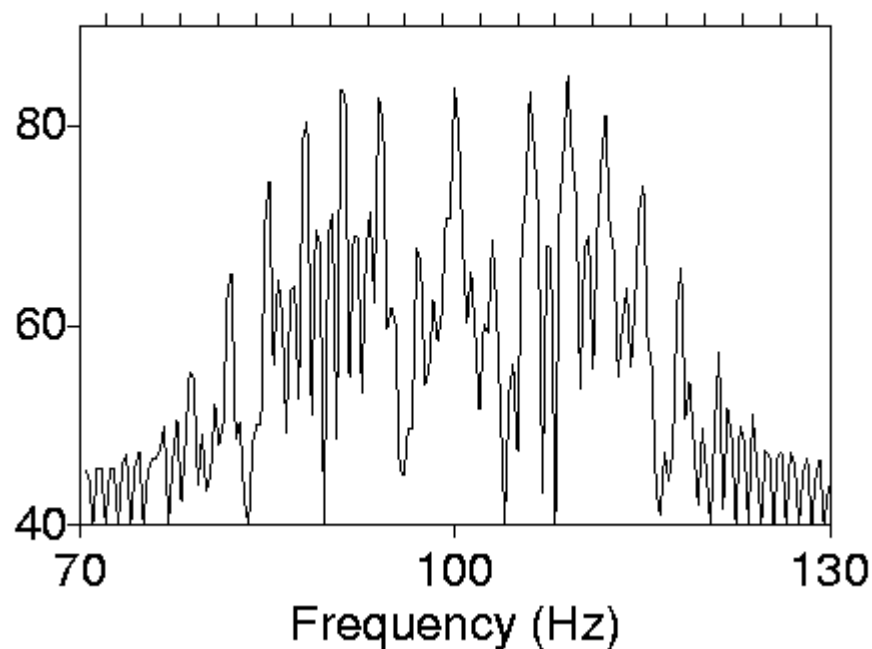
Example: tone with modulated frequency: modulation frequency 3 Hz

$$\sin(2\pi \cdot 100 \cdot x - 4 \cdot \cos(2\pi \cdot 3 \cdot x))$$

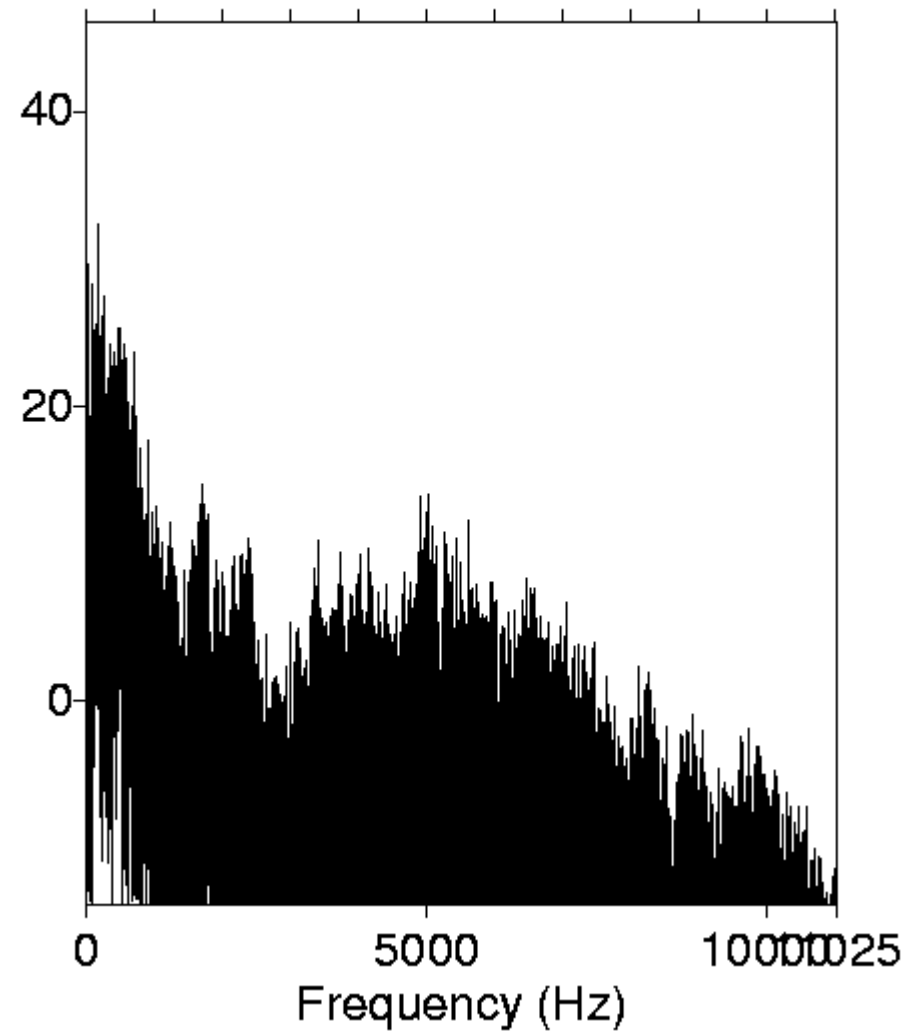
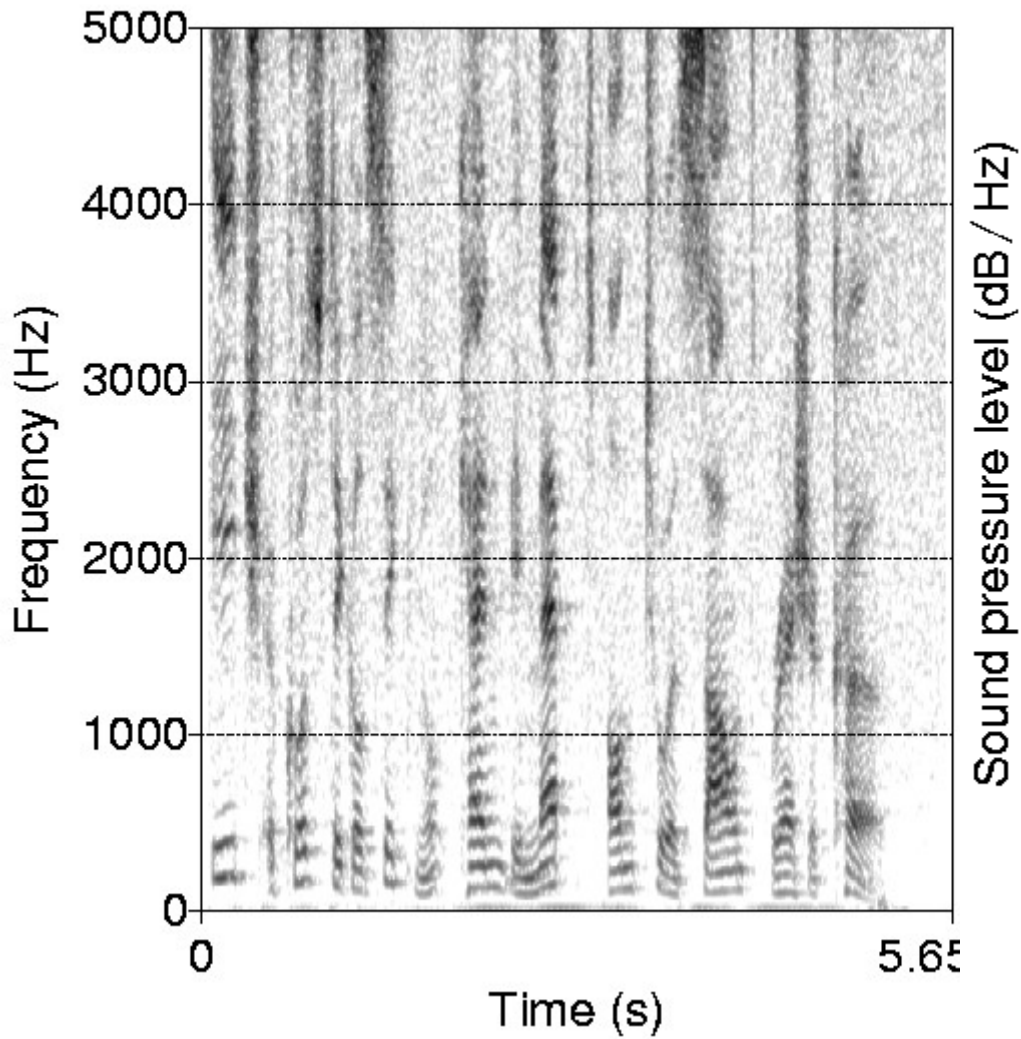
Spectrogram  $\Delta=0.05$  s



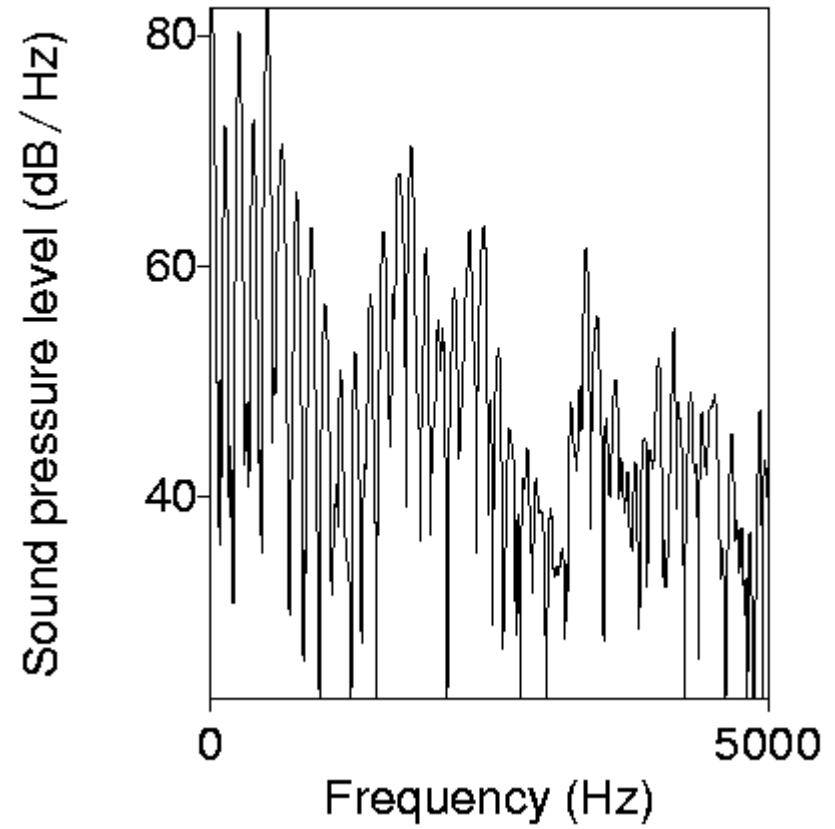
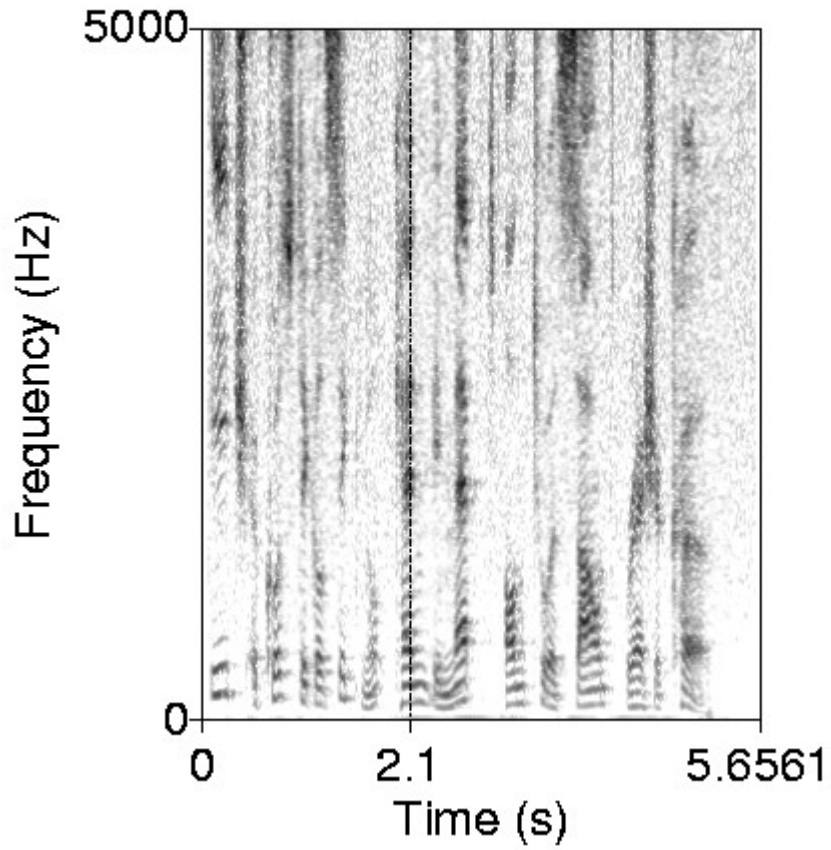
Spectrum



Spectrogram and spectrum of  
"Each acoustical signal can be mapped onto a sound pressure curve"



Spectrogram and spectrum taken at 2.1 s with gaussian window 0.05 s



# Wavelet analysis

wavelet analysis is a windowed Fourier analysis, where the time window  $g(t)$  depends on the frequency, such that the number of wiggles is always the same:

$$\tilde{p}(t, \nu) = \int dt' g((t - t')\nu, \nu) e^{i 2\pi \nu (t - t')} p(t')$$

$$\tilde{p}(t, \nu) = \int du h(u) p(u + \frac{t}{2\pi\nu})$$

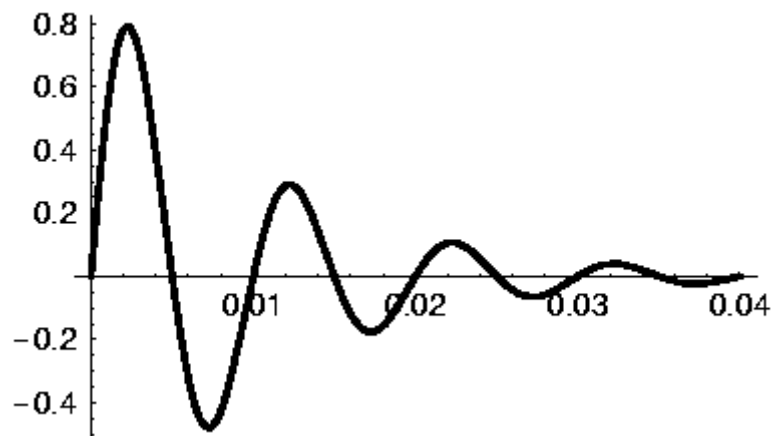
$$u = 2\pi\nu(t' - t), \quad h(u) = g(u) e^{iu}$$

# Examples for wavelet: Gaussian damped hamonics:

$$h(u) = e^{iu-u}$$

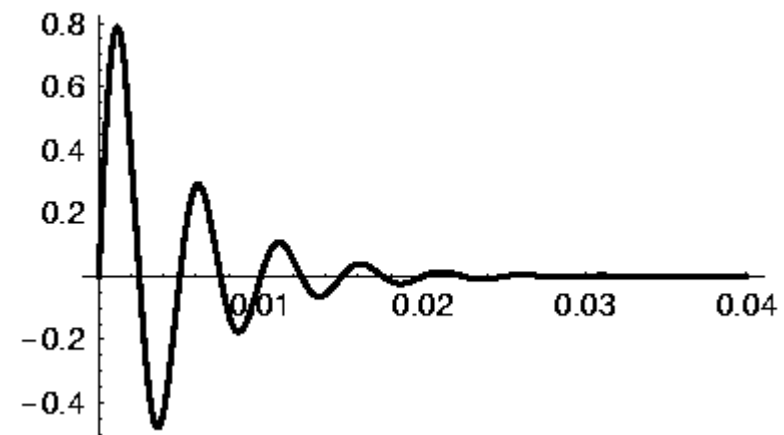
$$\nu = 100/(2\pi)$$

$\text{Im}h(u)$



$$\nu = 200/(2\pi)$$

$\text{Im}h(u)$



$h(u)$  is called a "note" :

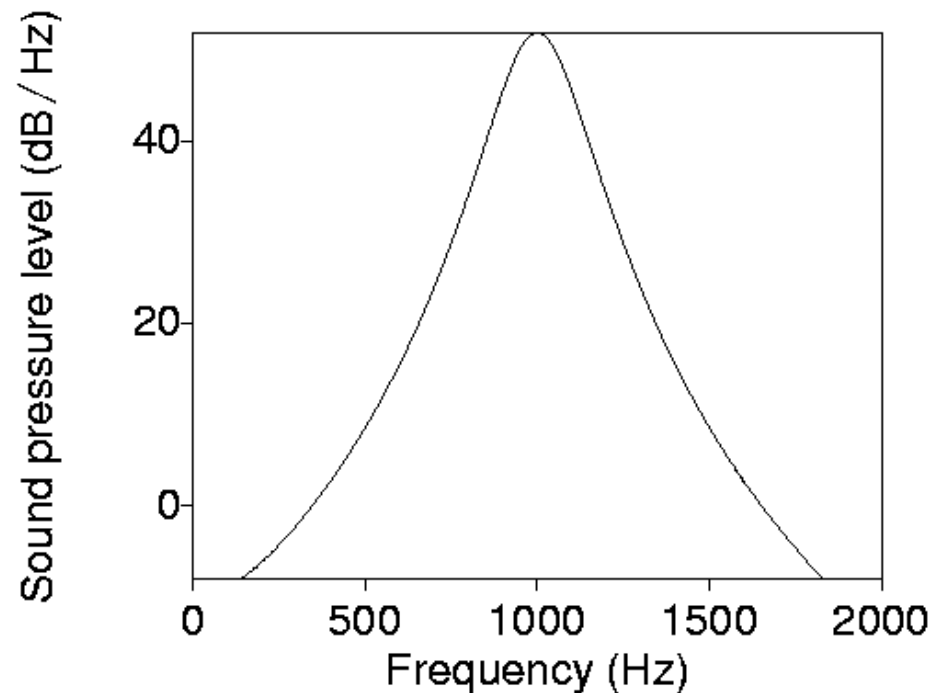
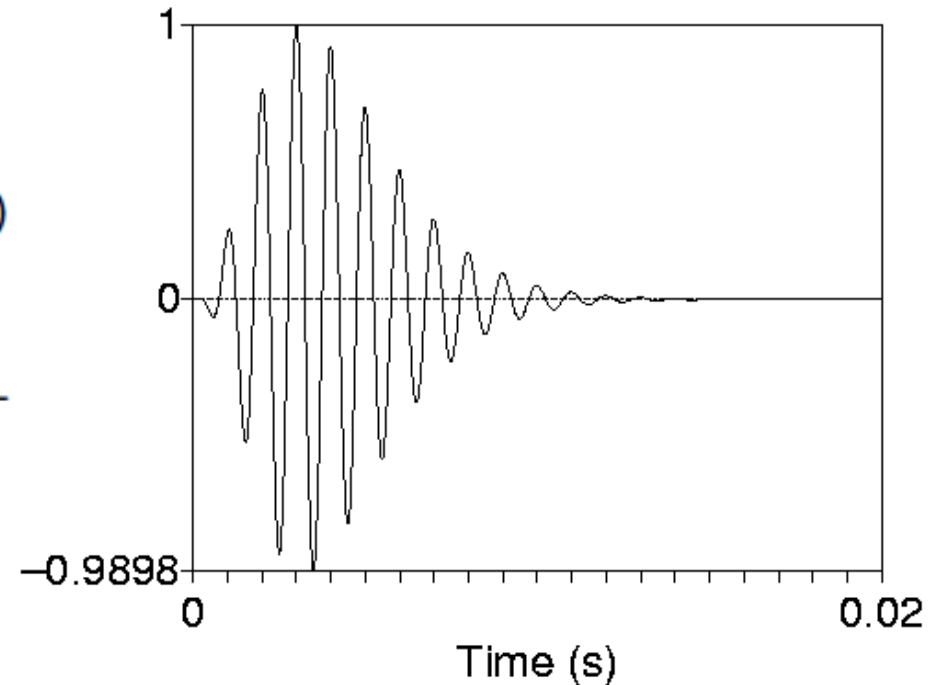
102praat note.wav

In acoustics the gamma-tone is a particularly popular wavelet

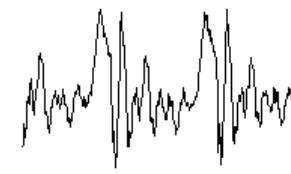
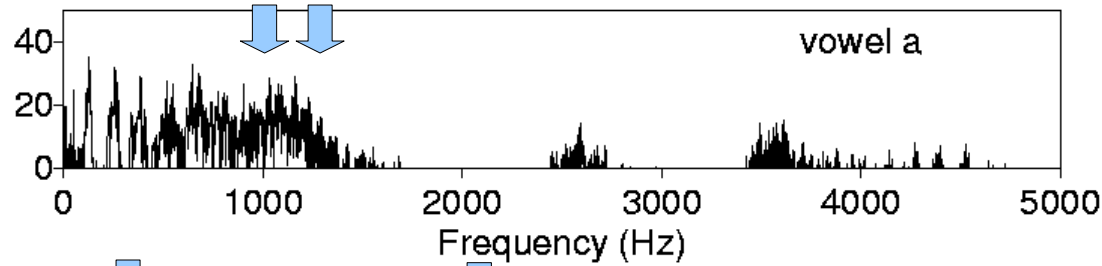
$$p_\gamma(t) = t^{\gamma-1} e^{-2\pi b(\nu_0)t} \cos(2\pi \nu_0 t)$$

$$\tilde{p}_\gamma(\nu) = \Gamma(\gamma + 1) \left( \frac{1}{(b(\nu_0) + i(\nu + \nu_0))^\gamma} + \frac{1}{(b(\nu_0) + i(\nu - \nu_0))^\gamma} \right)$$

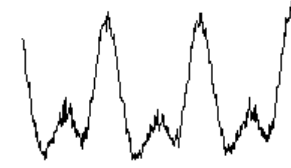
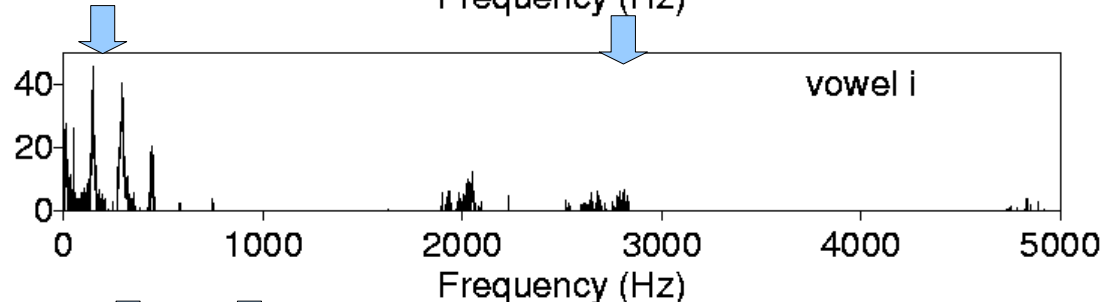
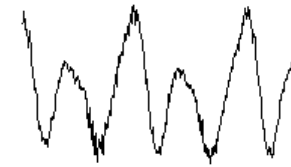
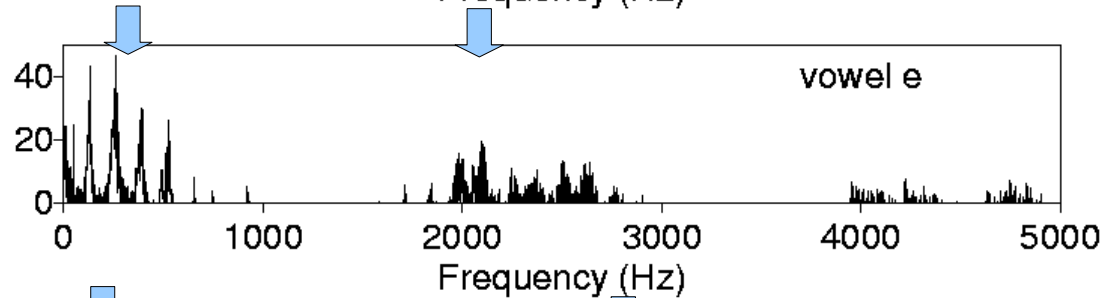
signal and spectrum of  
gamma-tone with  
t=4  
b = 150 Hz  
 $\nu_0 = 1000$  Hz



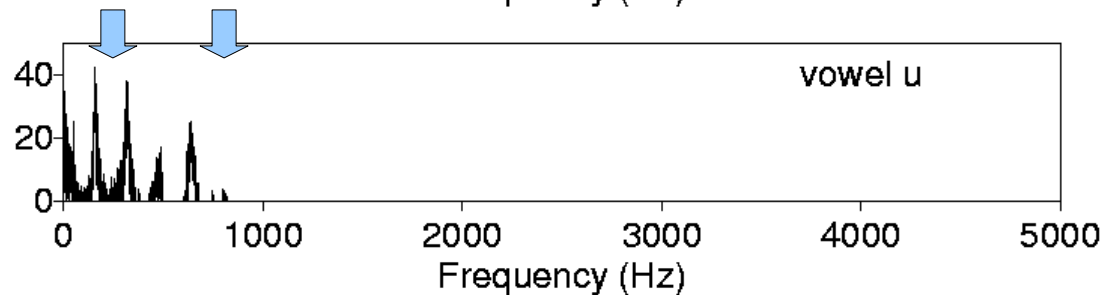
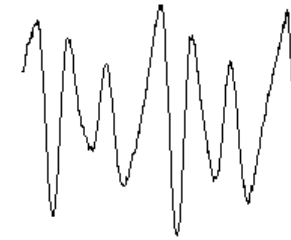
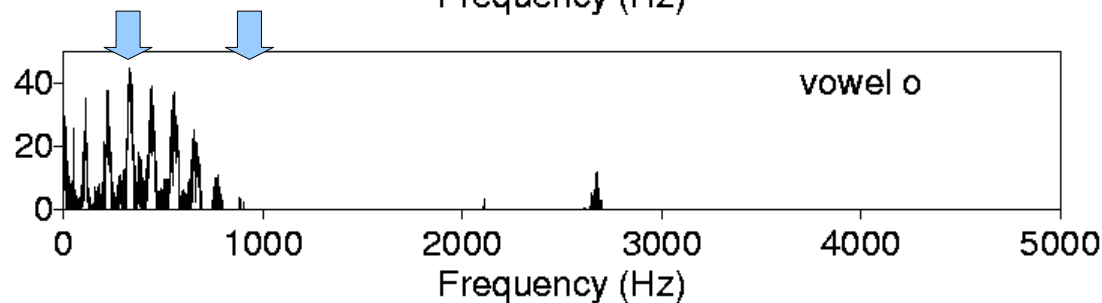
# Vowels, spectra and signals



102praat  
vokale-hgd  
rauschen fuer  
rauschvokale



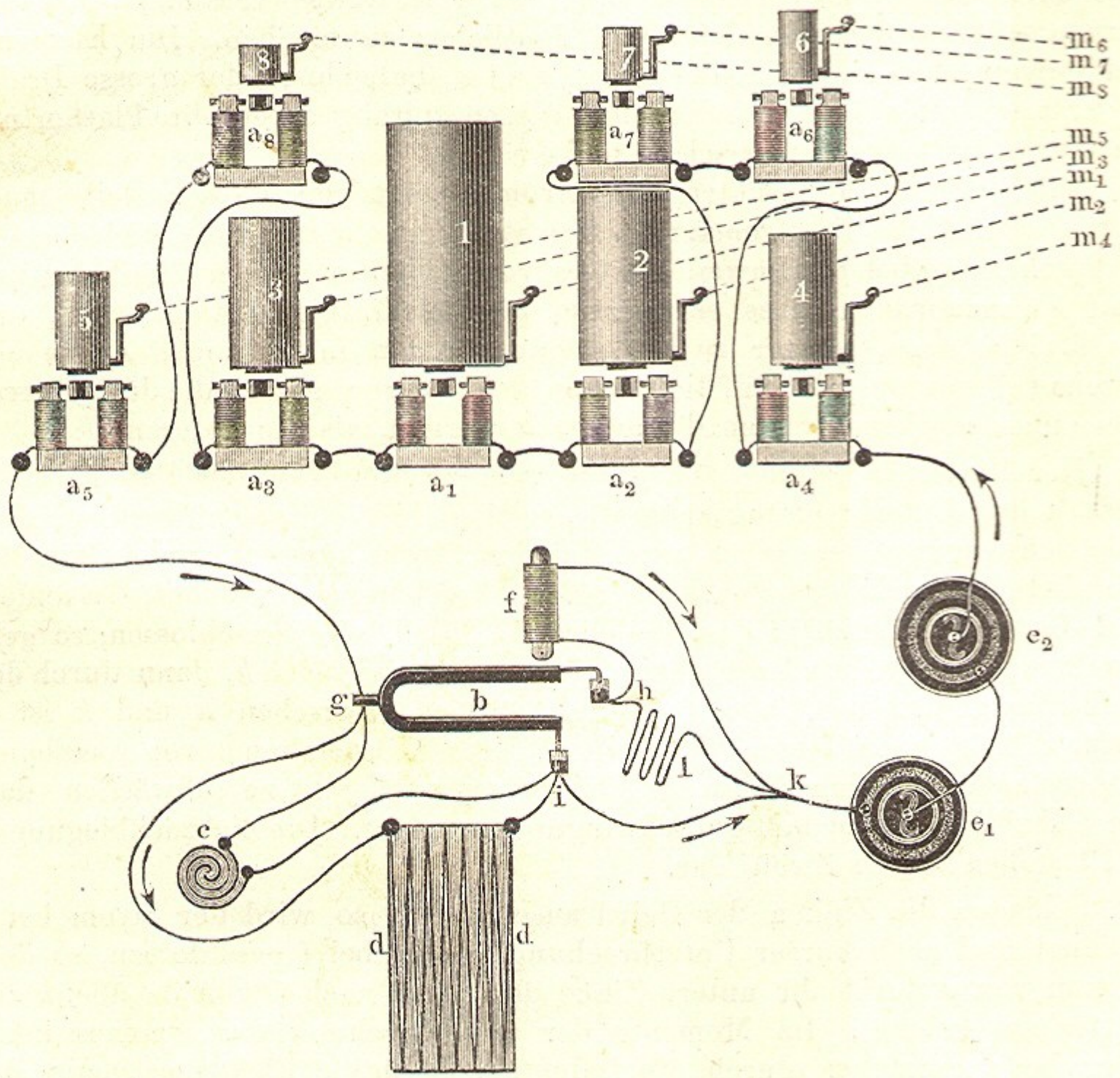
a	1000	1400
e	500	2300
i	320	3200
o	500	1000
u	320	800



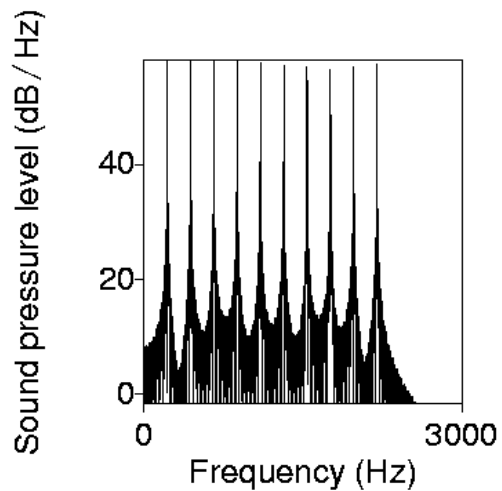
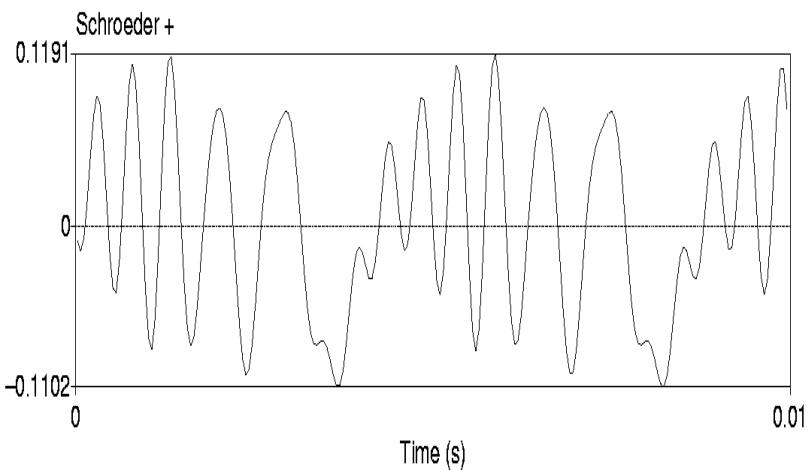
0.1      0.11      0.12



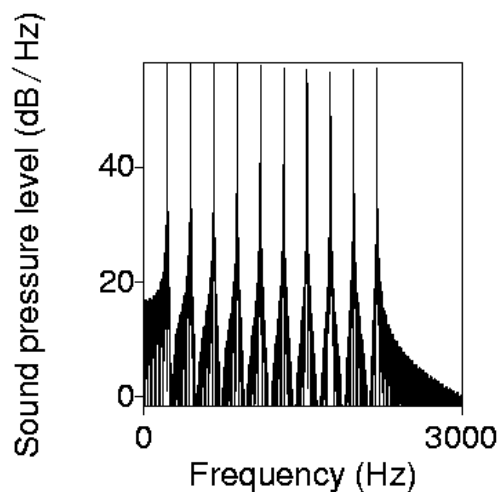
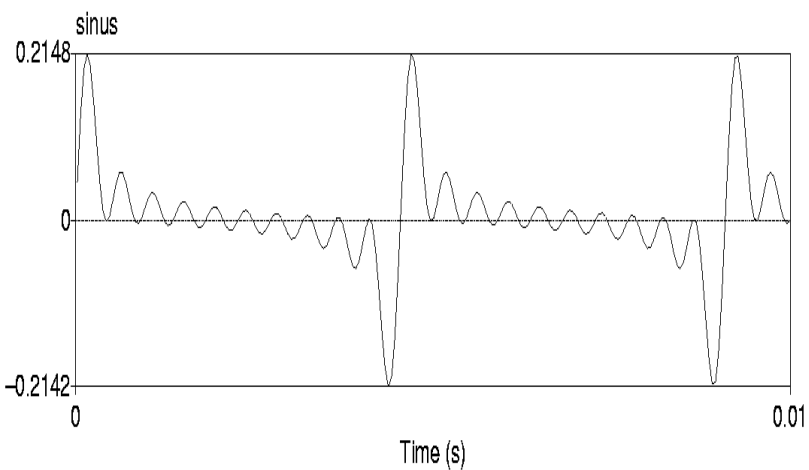
Fig. 64.



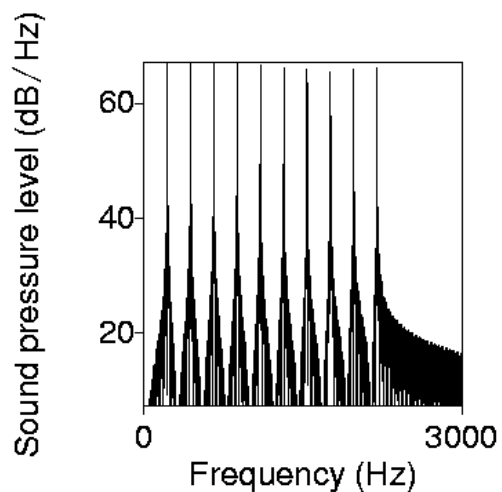
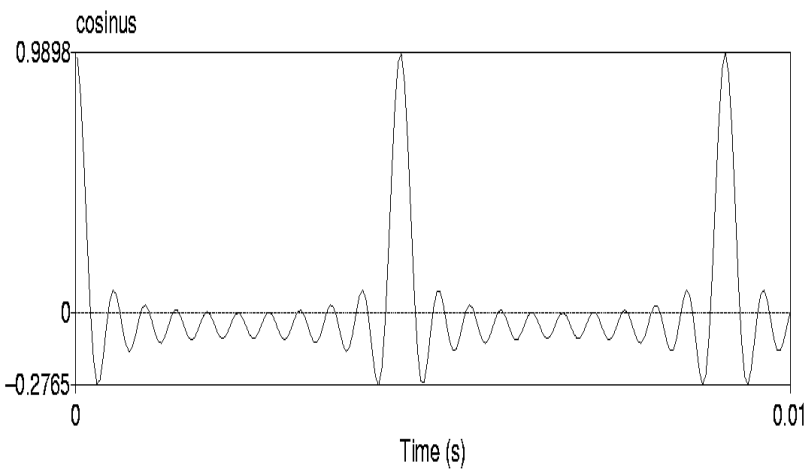
App. of  
Helmholtz  
to produce  
artificially  
vowels  
sponsored by  
Ludwig II von  
Bayern



The sound pressure curves look different, but the sounds are practically indistinguishable



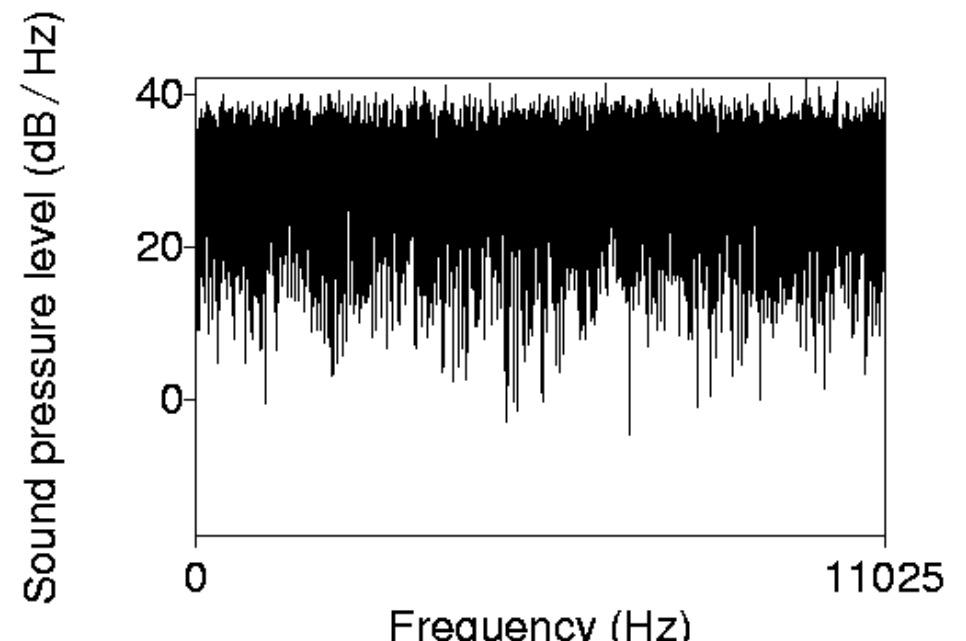
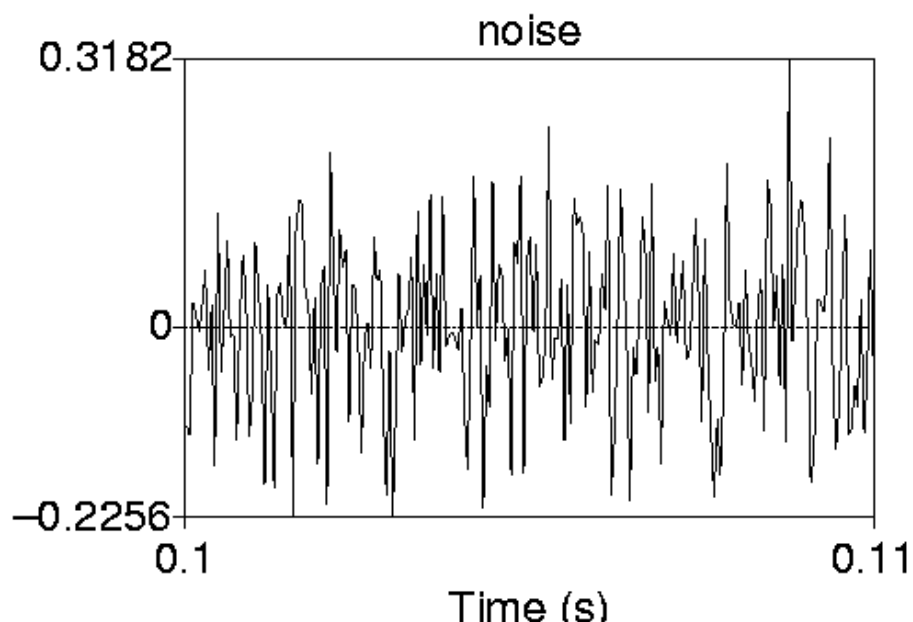
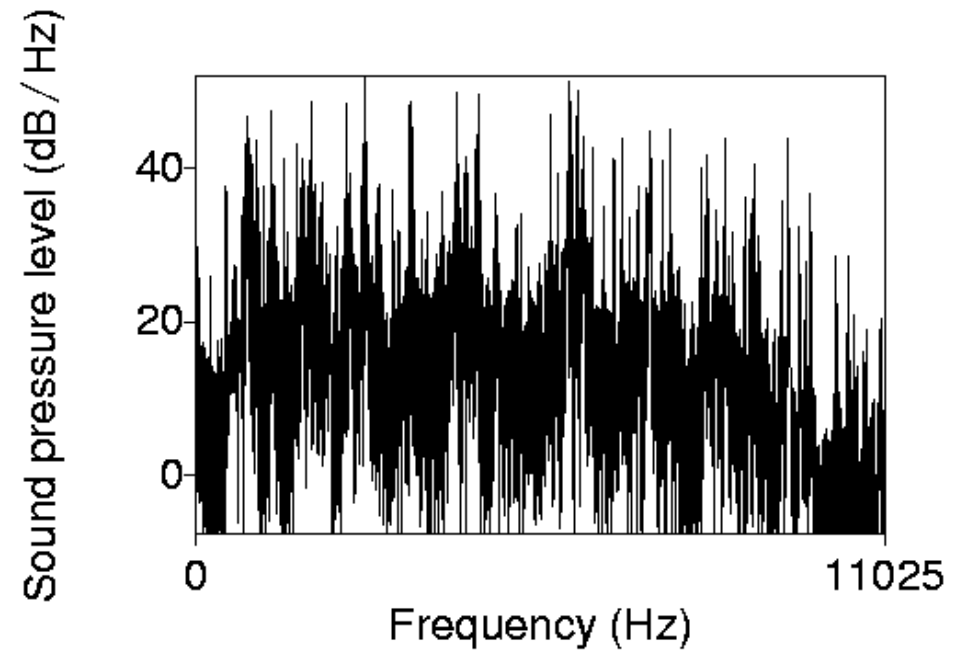
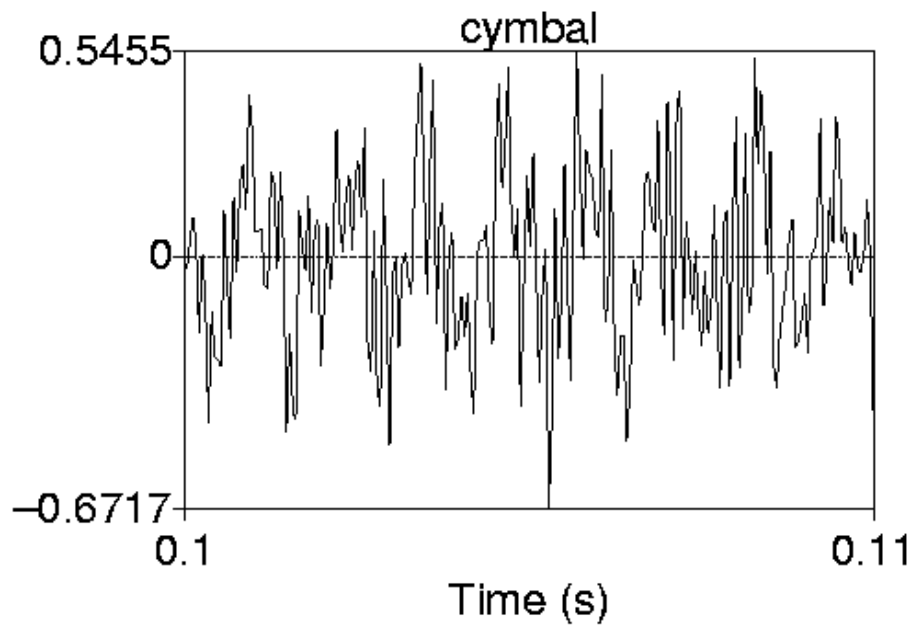
but the power spectra are up to a scale identical

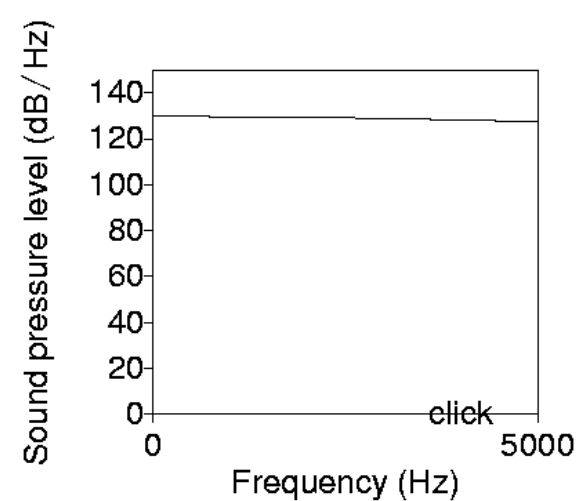
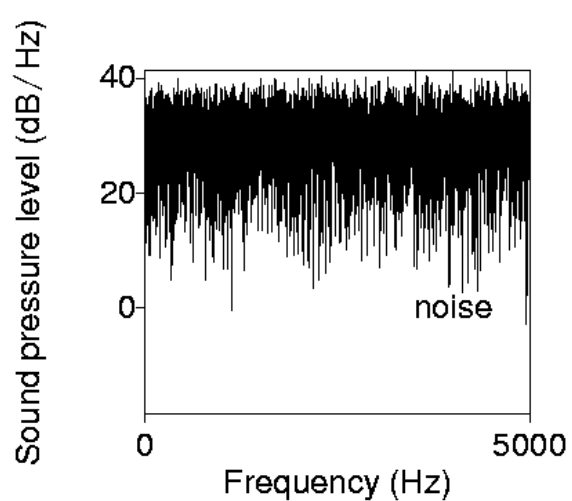
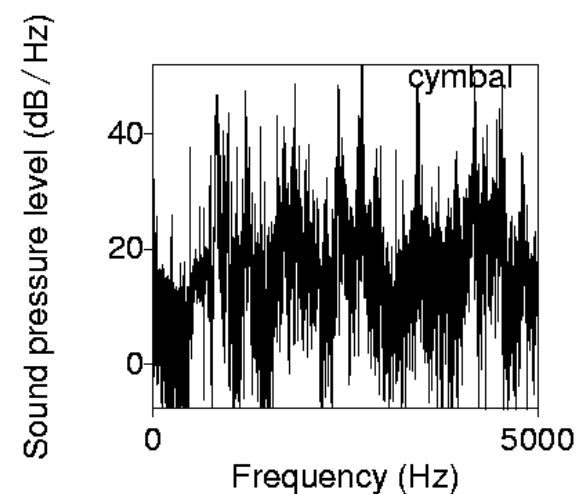
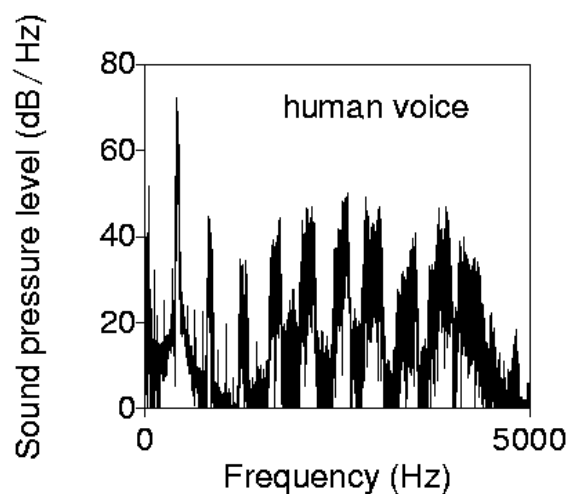
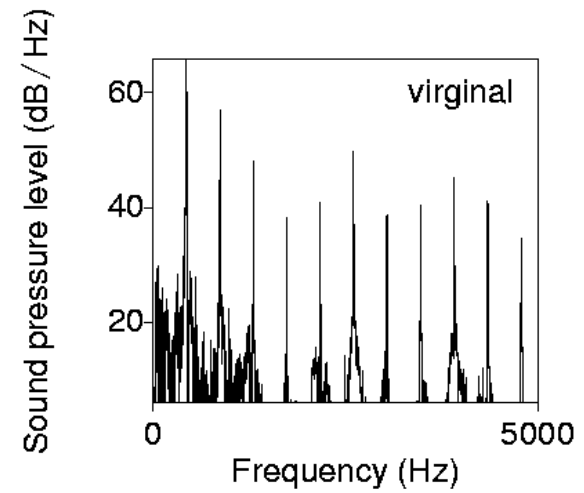
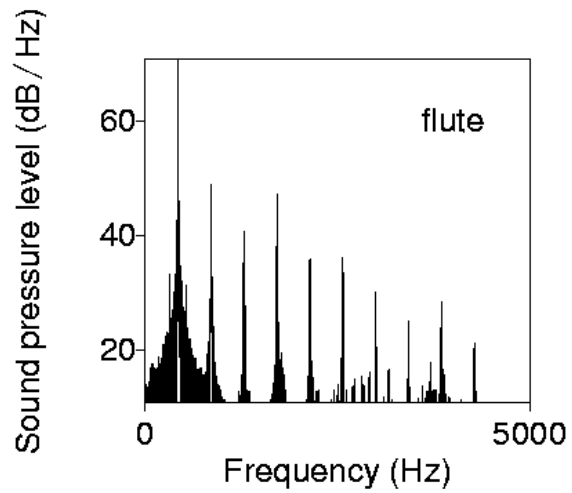


First refinement:

102praat cymbal.wav, noise.wav

Timbre determined by the **power spectrum**





Few spectral lines (few harmonics) soft tone  
 many harmonics sharp tone

102praat  
 flute.wav spinett.wav  
 i-reg.wav cymbal.wav  
 noise.wav click.wav

# Ohm's Law of Acoustics:

Ohm 1841, Helmholtz 1863

- 1) Ear performs a (windowed) Fourier analysis
- 2) Phases play no role.
- 3) The different Harmonics of a periodic tone are fused
- 4) Pitch of a sound determined by lowest harmonic,

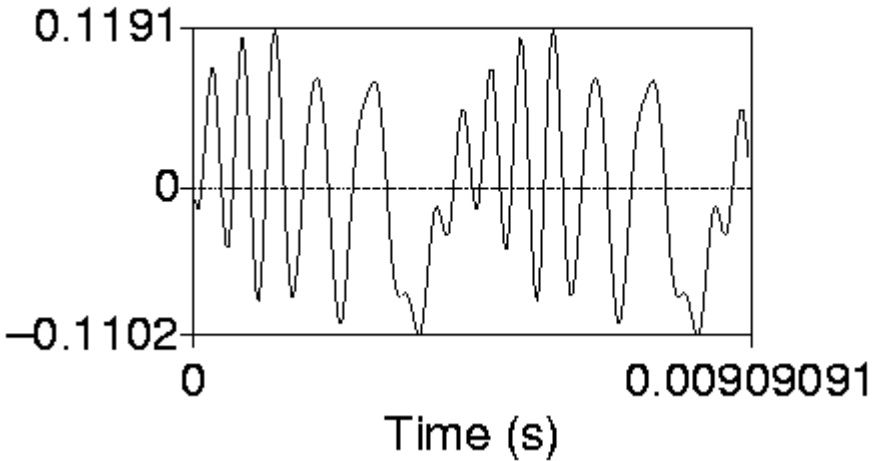
roughly:

long time scales ( $>0.1$  s) processed as temporal properties,

short time scales ( $< 0.1$  s) as power spectral properties

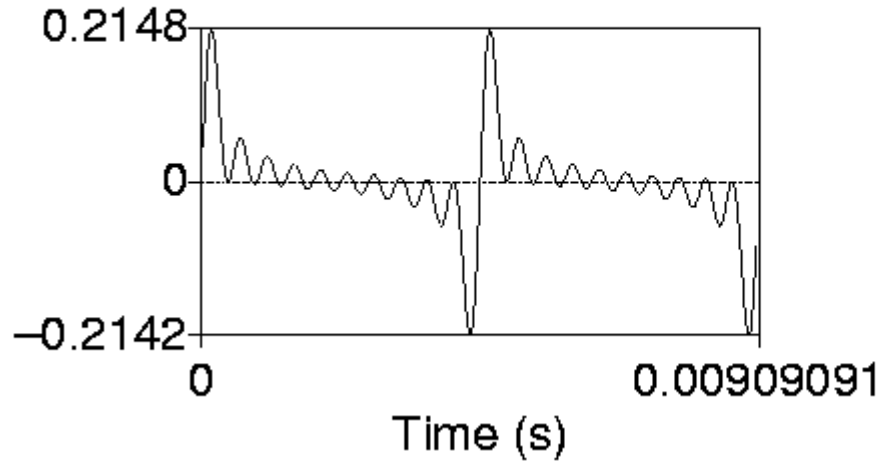


Independence of phases:



$$p(t) = \sum_{n=1}^{10} \sin(2\pi (n 220 t + \frac{n^2}{10}))$$

$$\tilde{p}(2\pi\nu) = \sum_{n=1}^{10} i\delta(\nu - n 220) e^{i 2\pi n^2 / 10}$$



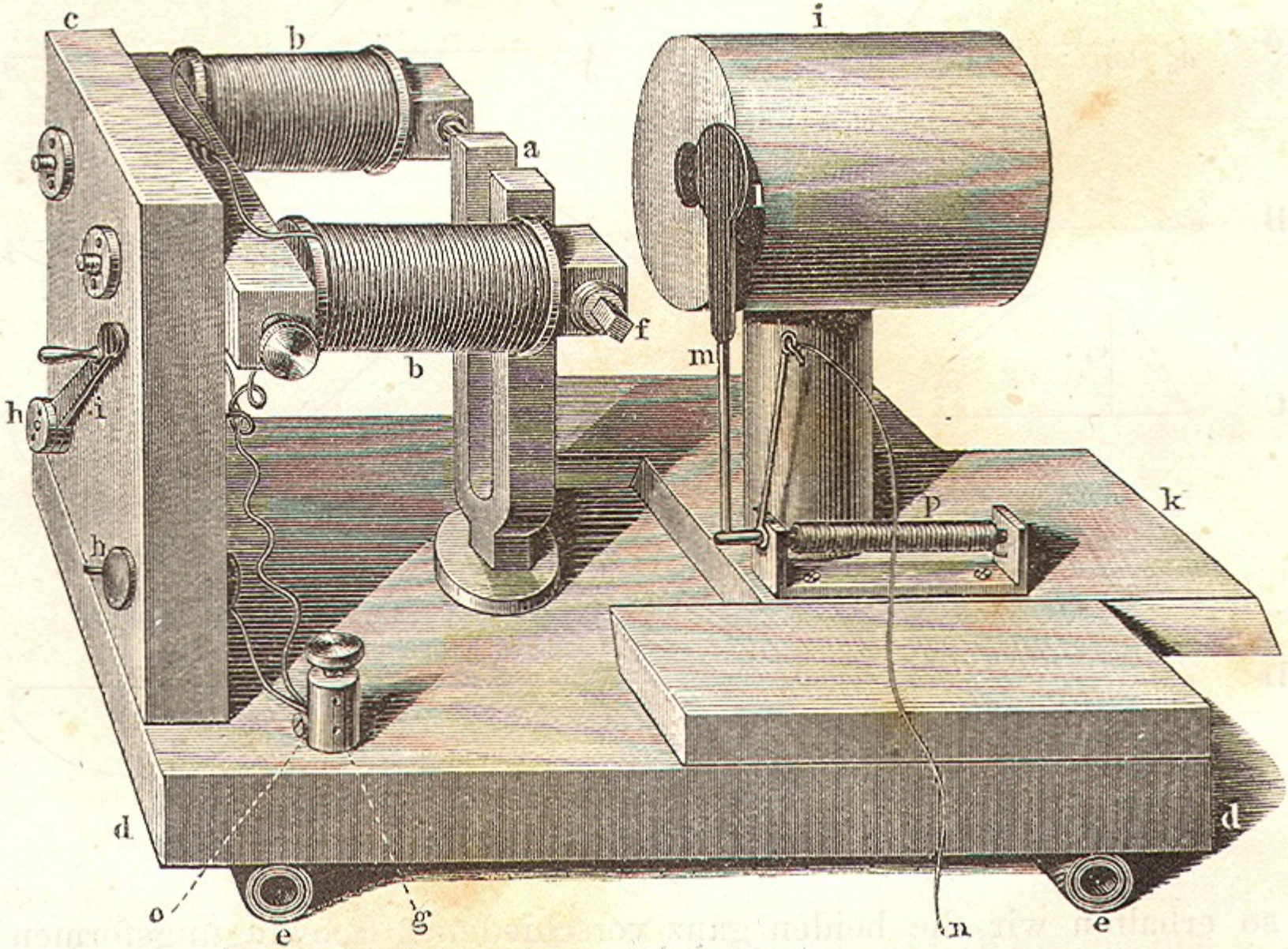
$$p(t) = \sum_{n=1}^{10} \sin(2\pi n 220 t)$$

$$\tilde{p}(2\pi\nu) = \sum_{n=1}^{10} i\delta(\nu - n 220)$$

schroederplus10, sin10

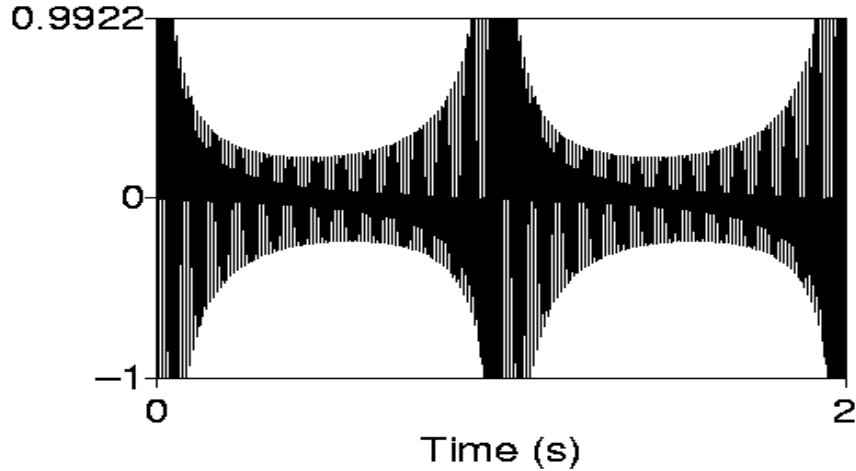
Apparatus of helmholtz to test the independence on phases (Tonempf.)

Fig. 32.

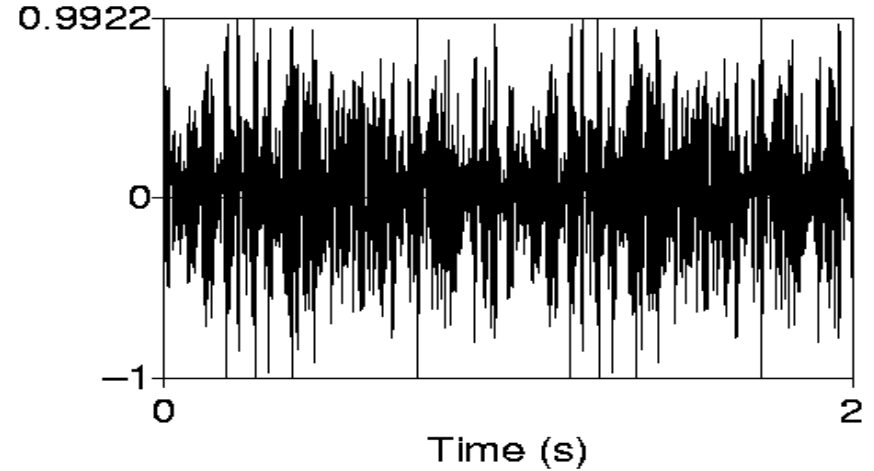


$$p(t) = \sum_{n=1}^{100} \sin(2\pi (100 + n) t)$$

$$p(t) = \sum_{n=1}^{100} \sin(2\pi (100 + n) t + 2\pi r[n])$$

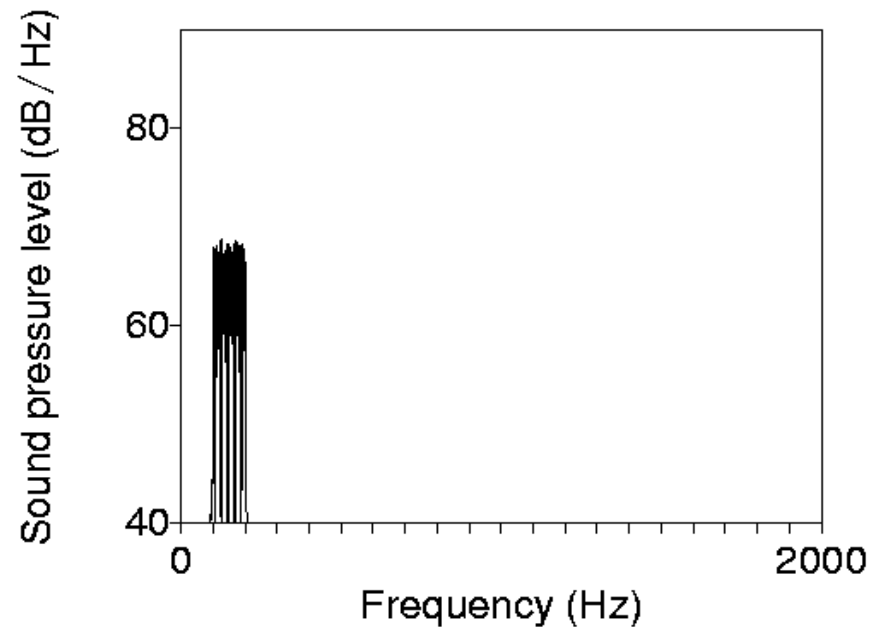
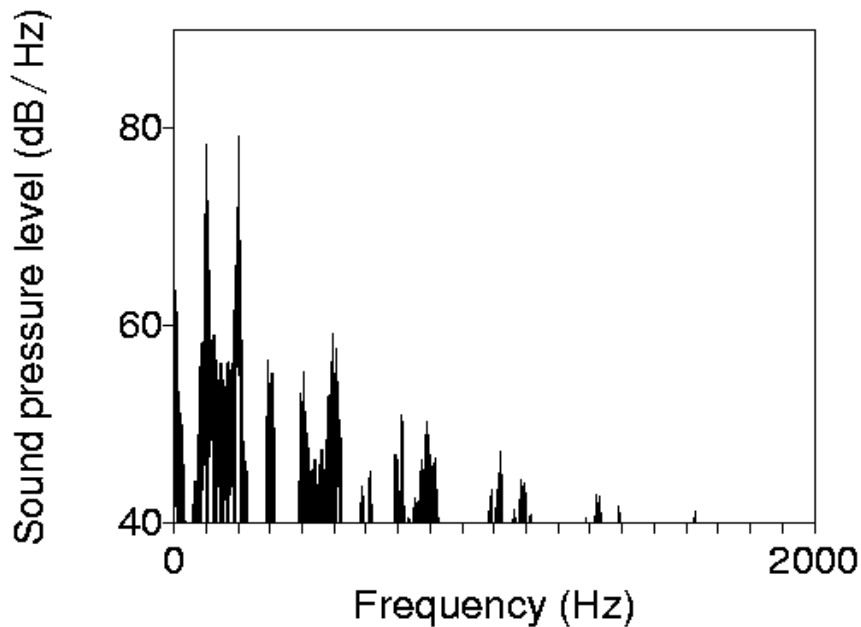


102praat: sin-coherent.wav



sin-random.wav

Fourier transform should only differ in phase, i.e. identical power spectrum  
 but: spectra over **finite time** interval look also different





Our ear performs a windowed Fourier analysis, since we can perceive directly time variations of ca 0.1 s, this should be the maximal width of the window.

Example beats and roughness.

$$\begin{aligned} e^{i2\pi\nu_1 t} + e^{i2\pi\nu_2 t} &= e^{i2\pi\frac{\nu_1+\nu_2}{2}t} \left( e^{i2\pi\frac{\nu_1-\nu_2}{2}t} + e^{-i2\pi\frac{\nu_1-\nu_2}{2}t} \right) \\ &= 2 \cos \left( 2\pi\frac{\nu_1-\nu_2}{2}t \right) e^{i2\pi\frac{\nu_1+\nu_2}{2}t} \end{aligned}$$

We expect:  $1/(\nu_1 - \nu_2) \ll 0.1$  s

two tones

$1/(\nu_1 - \nu_2) \gg 0.1$  s

beats

$\nu_1 = 2000; \nu_2 = 2500$

$\nu_1 = 2000; \nu_2 = 2010$

$\nu_1 = 2000; \nu_2 = 2120$

102praat  
beats-third.wav  
beats-10.wav  
beats-rough.wav

$1/(\nu_1 - \nu_2) > \sim 0.1$  s beats

$\nu_2/\nu_1 > \sim 1.2$  two tones

else rough. maximal roughness at  $\nu_2/\nu_1 \sim 1.06 - 1.12$

based on pysiology of the ear

### Theory of harmony Pythagoras - Helmholtz

	c(530)	530	1060	1590	2120	2650	3180	3710	4240
9:8	d(590)	590	1180	1770	2360	2950	3540	4130	4720
3:2	g(790)	790		1580		2370	3160	3950	
15:8	h(1000)		1000		2000		3000		4000
2:1	c(1060)		1060		2120		3180		4240

live-music

Fig. 60 A.

Calculations of Helmholtz for the degree of roughness for a violin

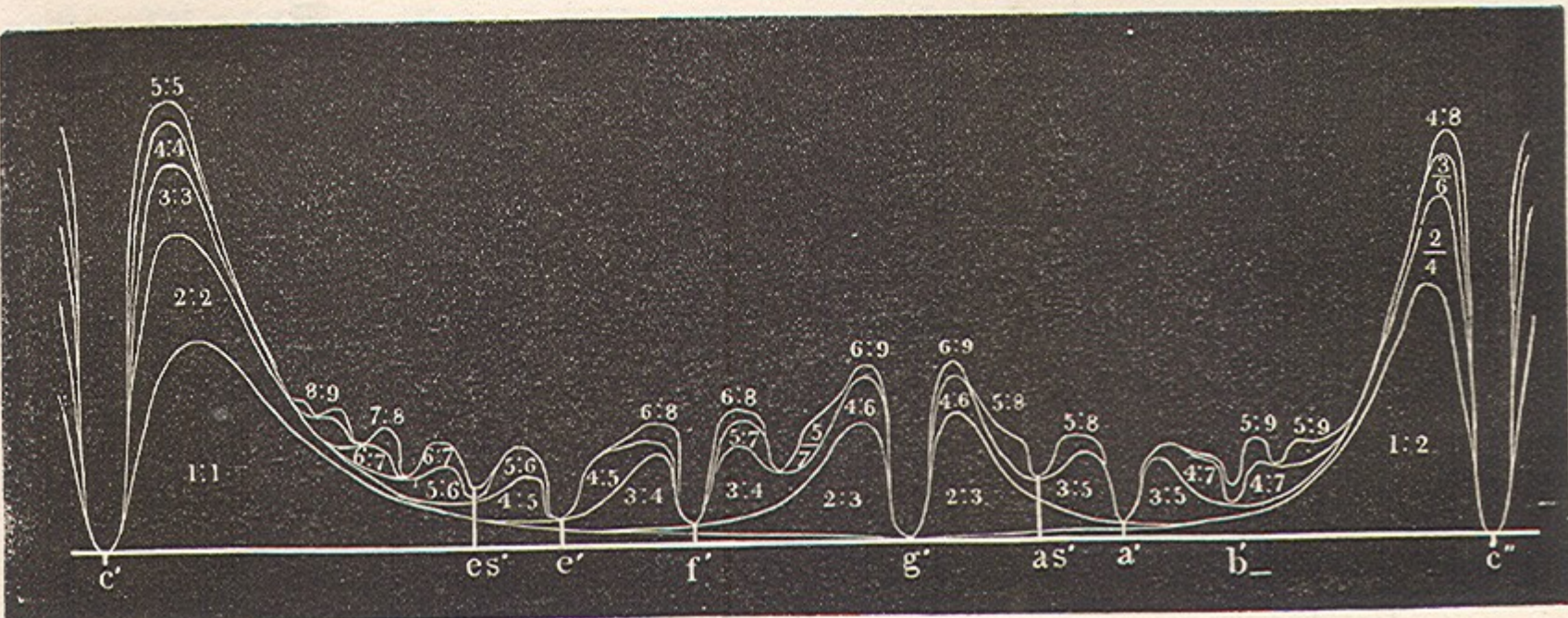
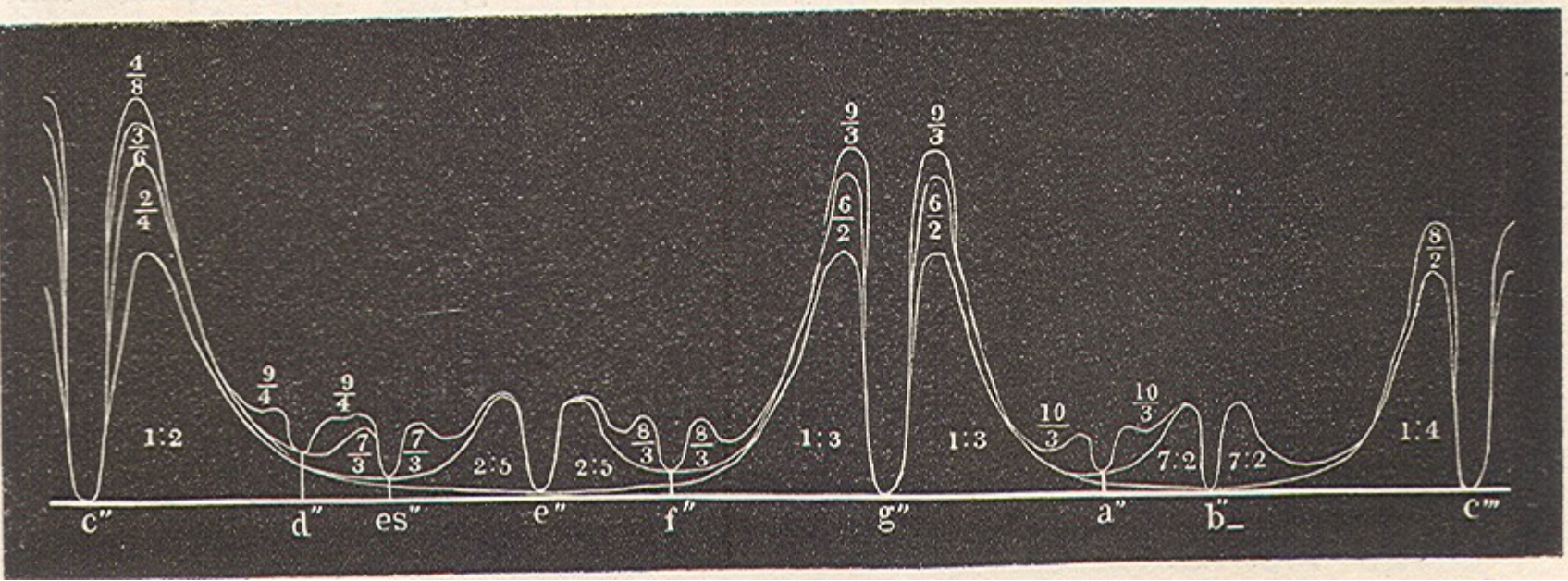
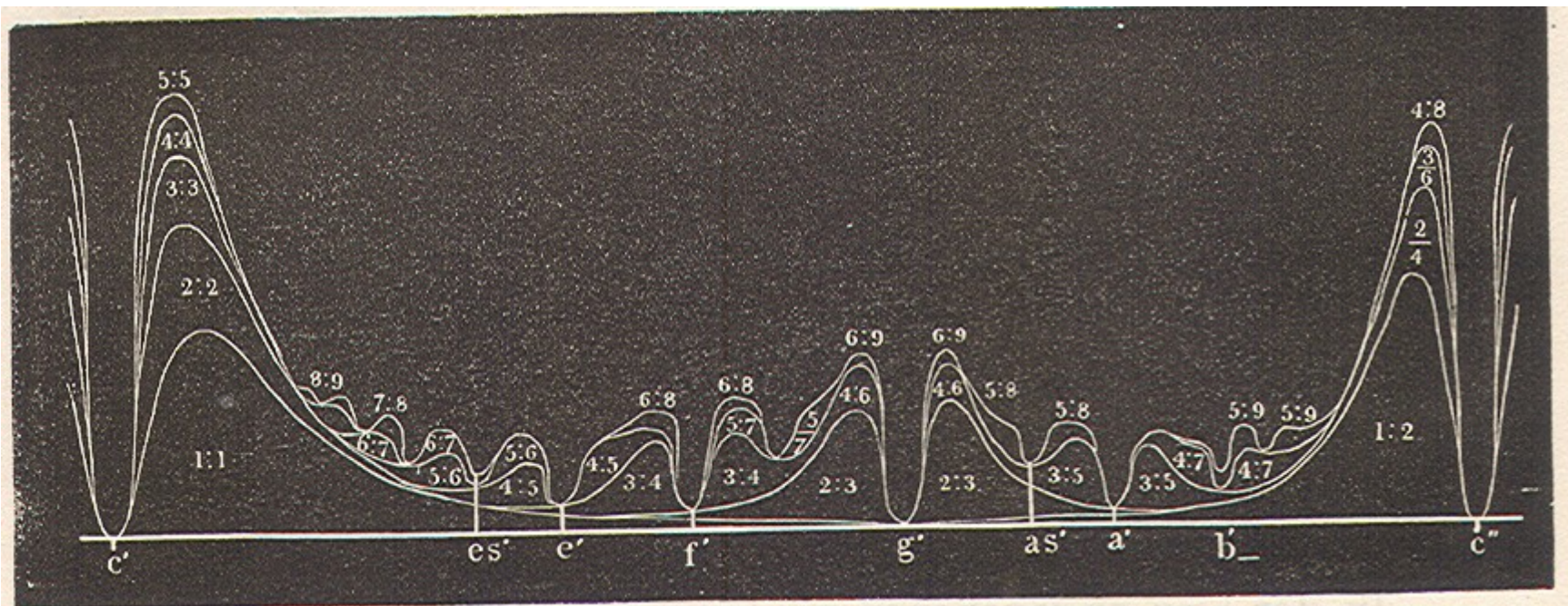


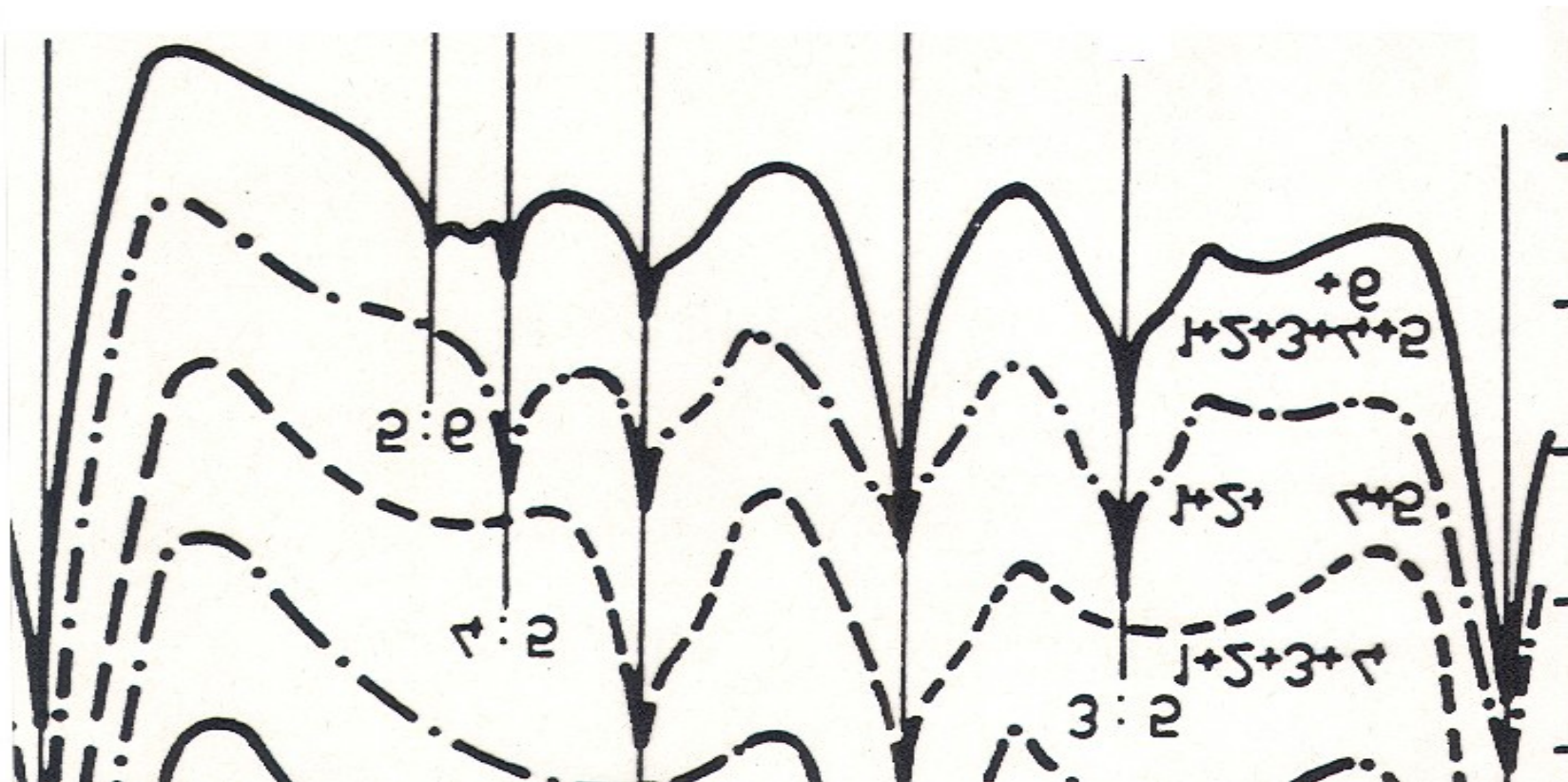
Fig. 60 B.





Helmholtz,  
1863

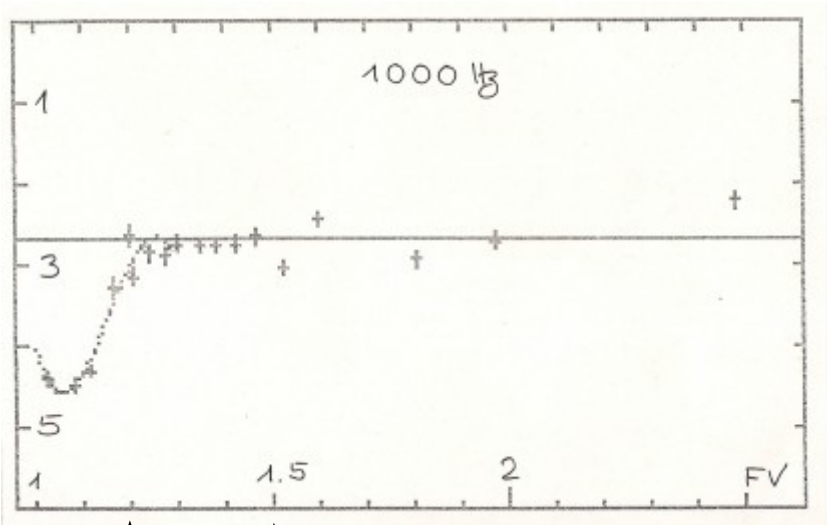
↑  
dissonanz



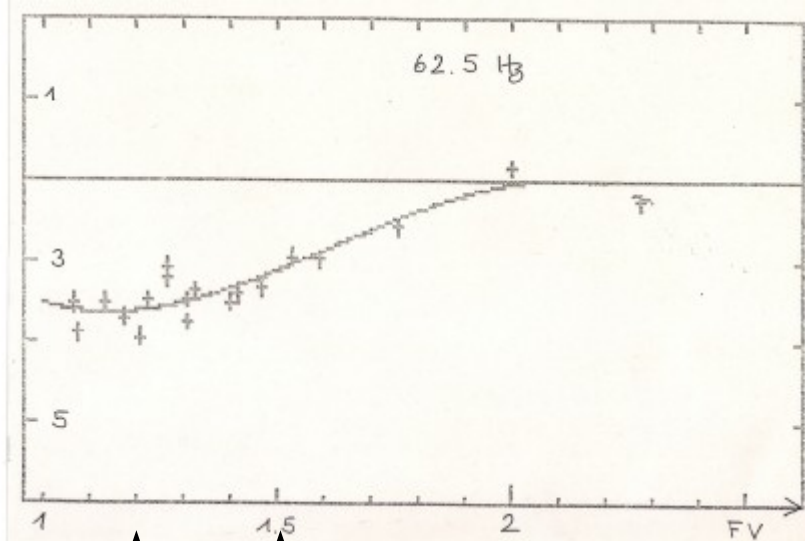
Kameoka,  
Kuriyagawa,  
1969

↑  
consonanz

Interval od roughness larger at low frequencies:



minor third  
fifth



minor third  
fifth

Dosch and Specht, 1986  
average of 89 subjects

Chopin, Scherzo

Musical score for Chopin's Scherzo. The score is written for piano and consists of two staves. The key signature has three flats (B-flat, E-flat, A-flat), and the time signature is 3/4. The piece is marked with a tempo of 'Allegretto'. The score includes various musical notations such as dynamics (ff, sf), articulation (accents), and fingerings (8, 5, 4, 1). There are also some handwritten annotations like 'Led.' and asterisks. The score is divided into measures by vertical bar lines, and there are some markings above the staff, possibly indicating fingerings or articulation points.

Beethoven, Appassionata

Musical score for Beethoven's Appassionata. The score is written for piano and consists of two staves. The key signature has three flats (B-flat, E-flat, A-flat), and the time signature is 3/4. The piece is marked with a tempo of 'Allegro molto'. The score includes various musical notations such as slurs, dynamic markings (f), and articulation. The score is divided into measures by vertical bar lines, and there are some markings above the staff, possibly indicating fingerings or articulation points.

Example: “pure tone”  $p(t) = \sin(2\pi 220 t)$

complex tone  $p(t) = \sum_{n=1}^5 \sin(2\pi n 220 t)$

But one can also hear the components, some people better, some worse.



# Fusion of tones

A tone with several harmonics: is it a single tone or a collection of tones?

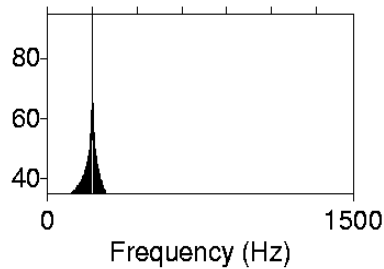
both:

Helmholtz:

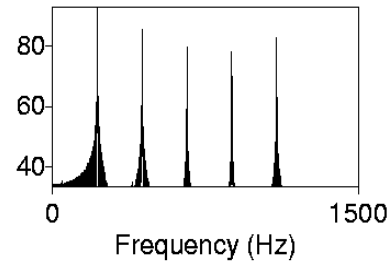
we perceive synthetically (perzipieren) the whole tone

we can perceive analytically (apperzipieren) the components of the tone.

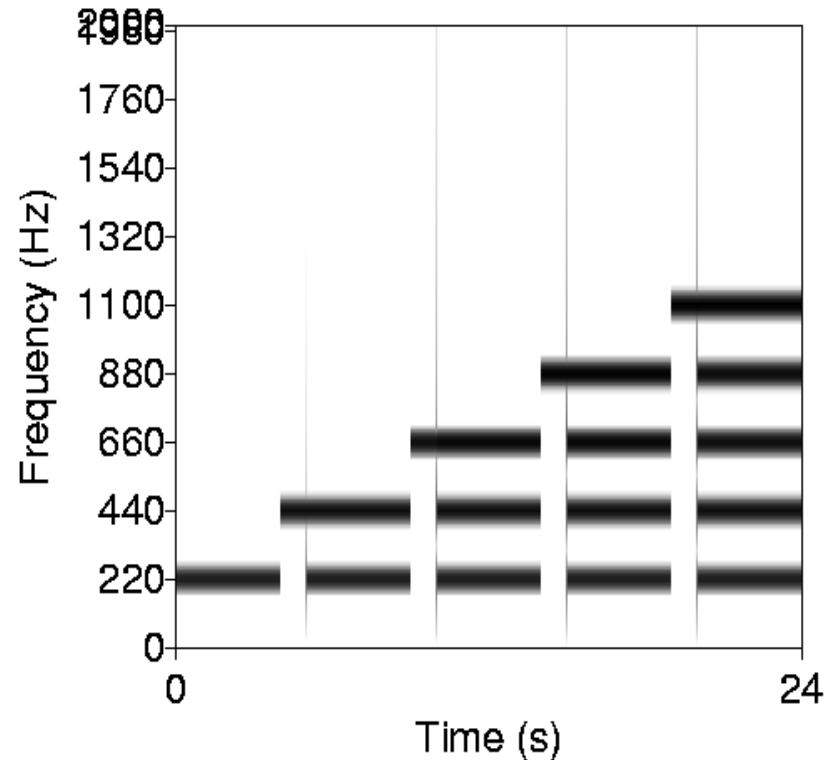
sound pressure level (dB/H):



sound pressure level (dB/H):



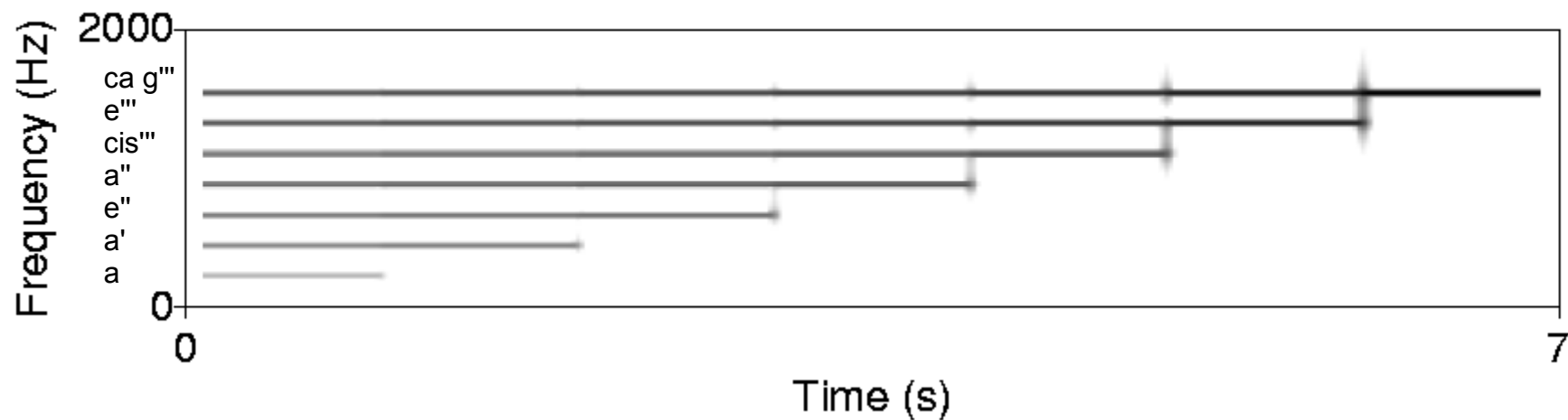
102praat  
sin-1.wav  
sin-5.wav  
chain-harmonic-insert.wav





The missing fundamental  
We now take off components from below

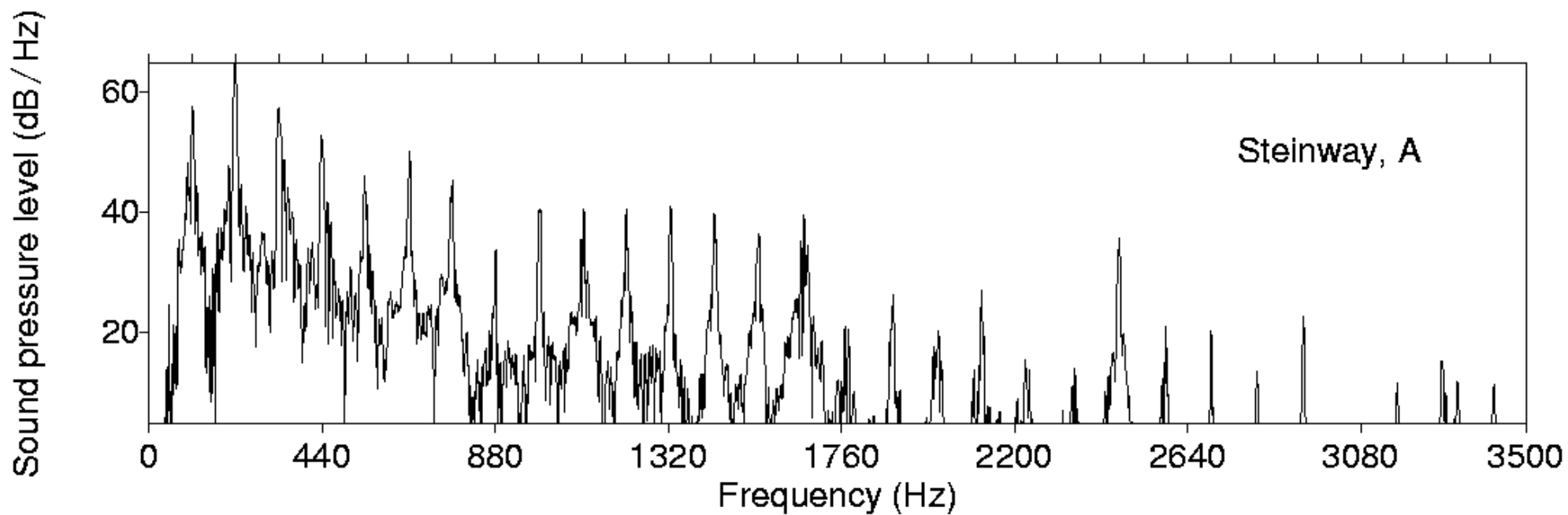
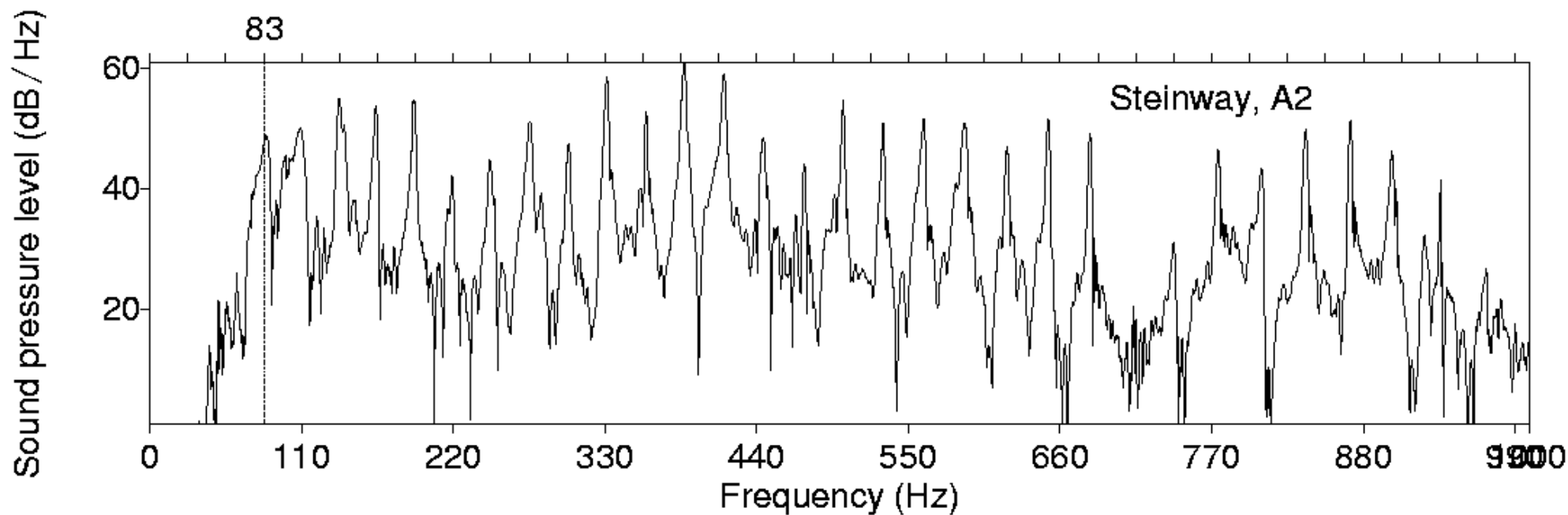
102praat chain-fundamental-tracking.collection



We take subsequently one harmonic tone out of the complex tone

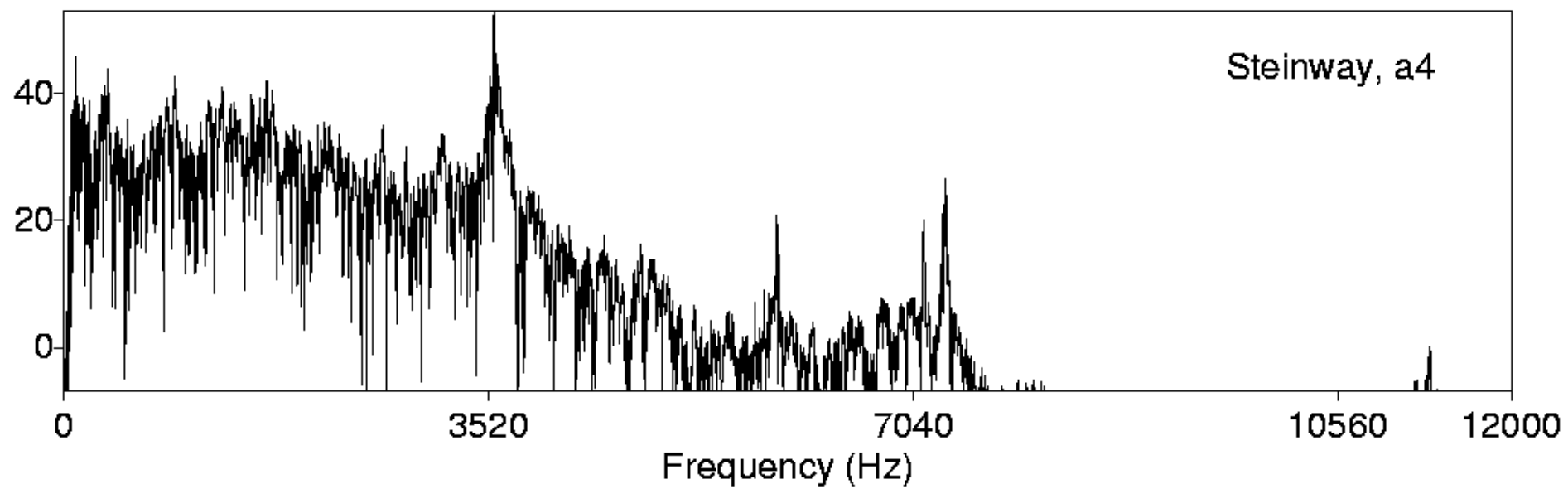
This happens also in musical instruments

103praat 27\_5Hz\_A2\_real\_steinway.wav 110Hz-A\_real



103praat 3250Hz-a4-real-steinway

Sound pressure level (dB / Hz)



## Explanation until ca 1940 : Difference tones

If a sound is processed linearly, nothing happens to the spectrum, except linear scaling.

But if the function is not linear, we observe additional spectral lines (partial tones)

$$(2a \cos(\omega_1 t) + 2b \cos(\omega_2 t))^n$$

$$= \left( a(e^{i\omega_1 t} + e^{-i\omega_1 t}) + b(e^{i\omega_2 t} + e^{-i\omega_2 t}) \right)^n$$

$$= (ab)^n (e^{ni\omega_1 t} + e^{i((n-1)\omega_1 + \omega_2)t} + \dots + e^{i((n-1)\omega_1 - \omega_2)t} +$$

...

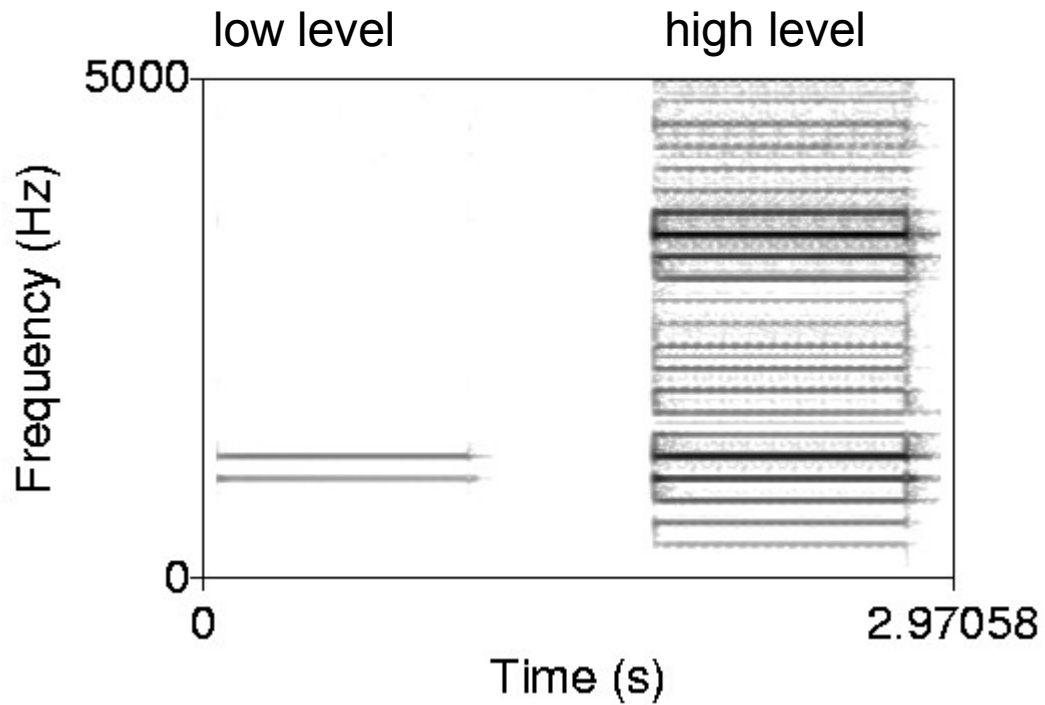
$$n = 2 : 2\omega_1, \omega_1 + \omega_2, \omega_1 - \omega_2, 2\omega_2$$

$$n = 3 : 3\omega_1, 2\omega_1 - \omega_2, \dots$$

For a harmonic tone  $p(t) = \sum_k^n \cos(2\pi k\nu_0 t)$

the difference tone  $\omega_{k+1} - \omega_k = 2\pi(k+1)\nu_0 - 2\pi k\nu_0 = 2\pi\nu_0$  is the fundamental tone

103praat dif-tone-laut, leise 1000,1220  
Hz

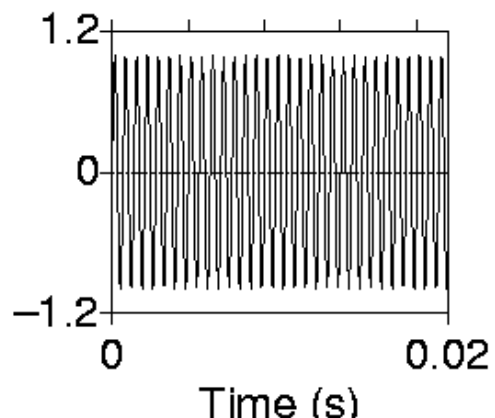
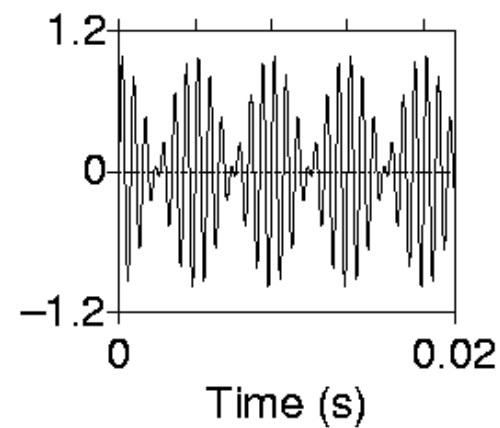
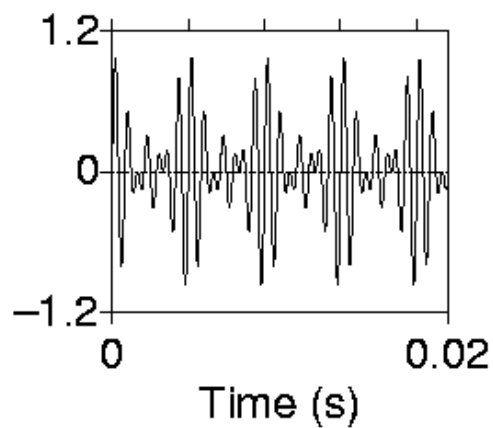
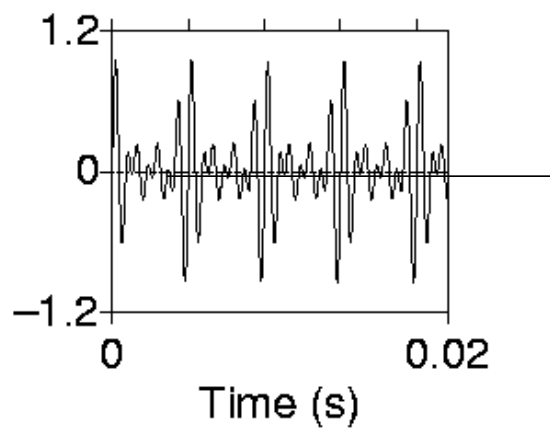
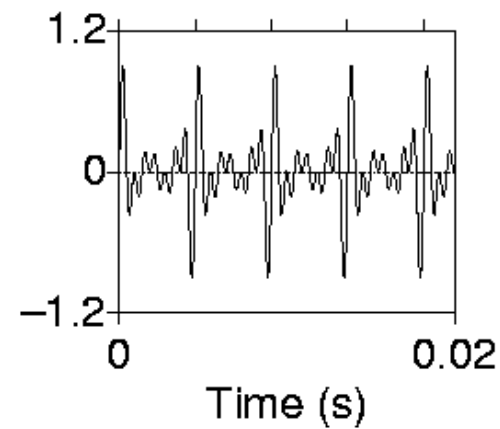
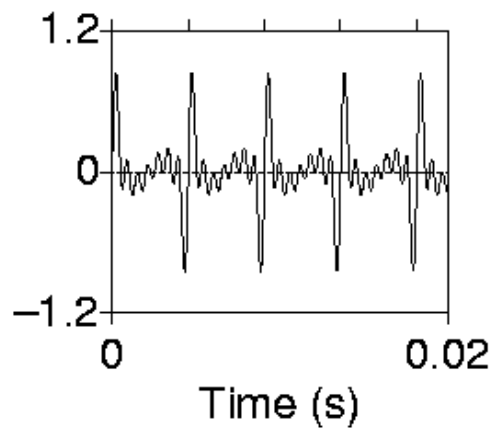
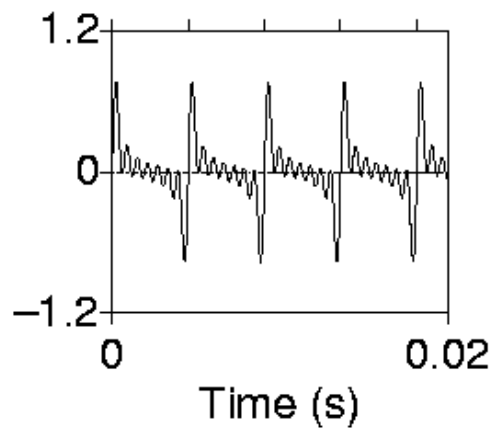


Spectrogram of the recorded tone, consisting of two components at 1000 and 1220 Hz. At the high level the nonlinear distortions are clearly visible.

Schouten:

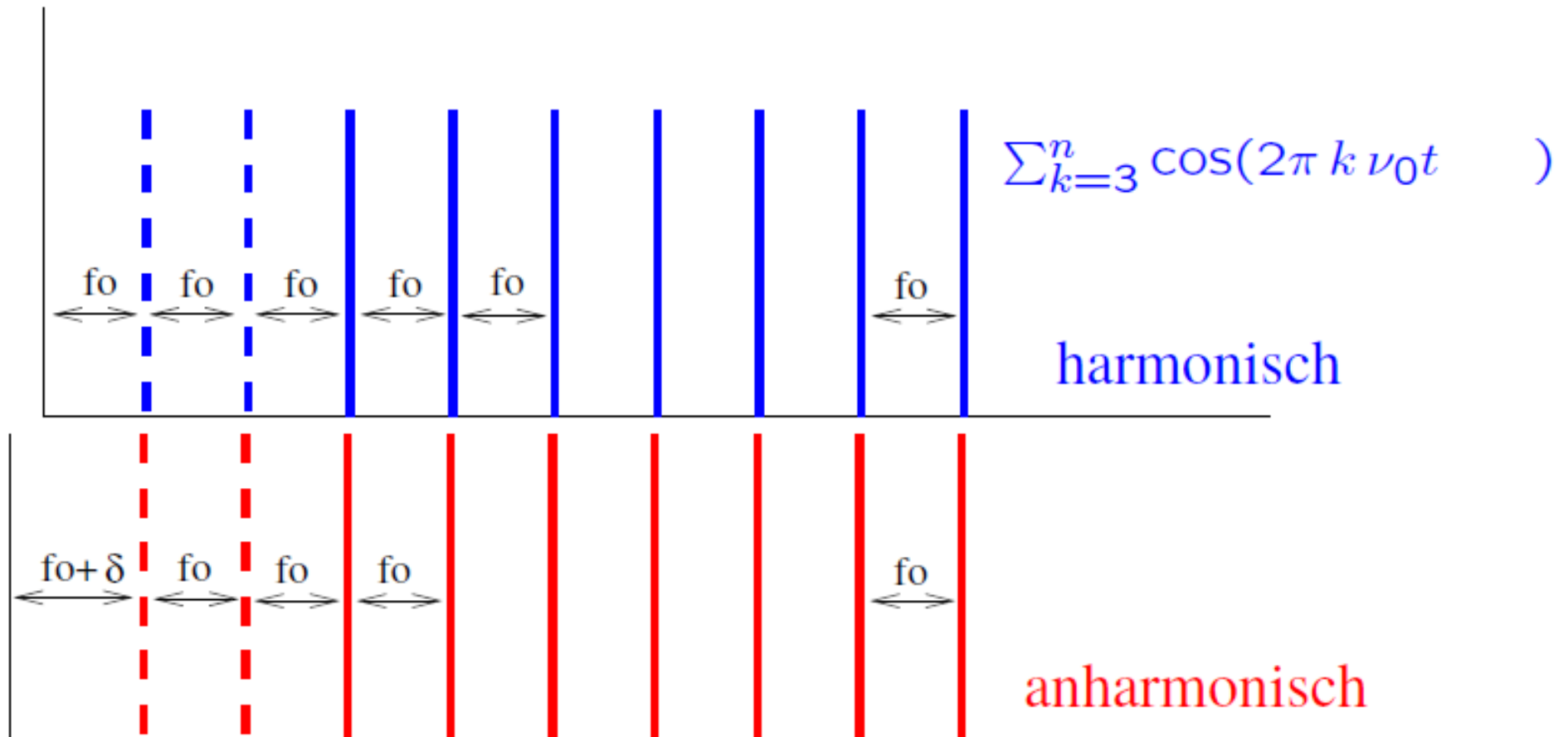
The periodicity of the tone is essential for the pitch of the "residue", that is the perceived, but not present fundamental tone.

Indeed sp curves show this periodicity:



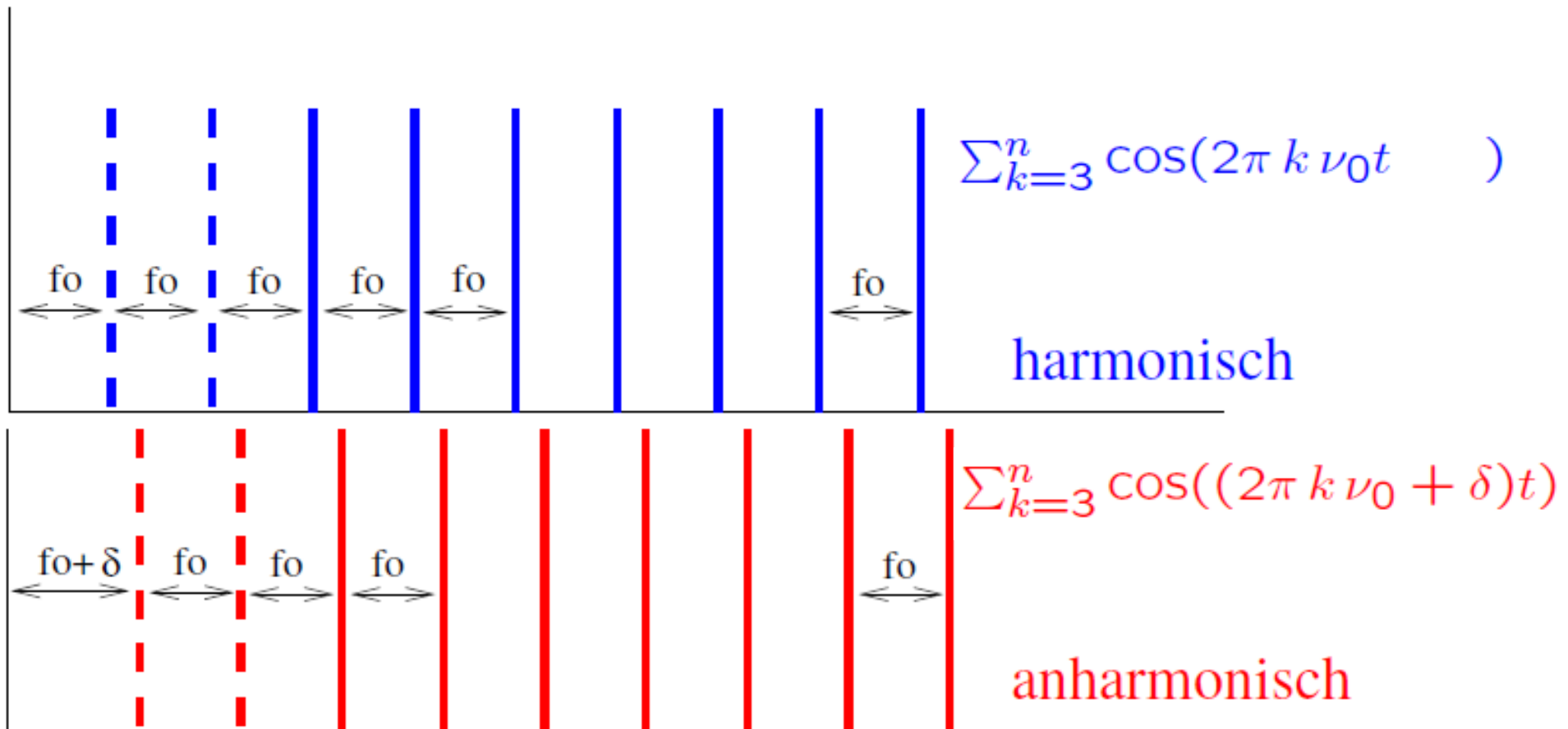
sp for the  
harmonic tones  
1-7, 2-7, 3-7,  
4-7, 5-7, 7-7,  
7

The difference tone was excluded by van Schouten by a series of ingenious experiments. A particular simple one is the shifted harmonic tone.





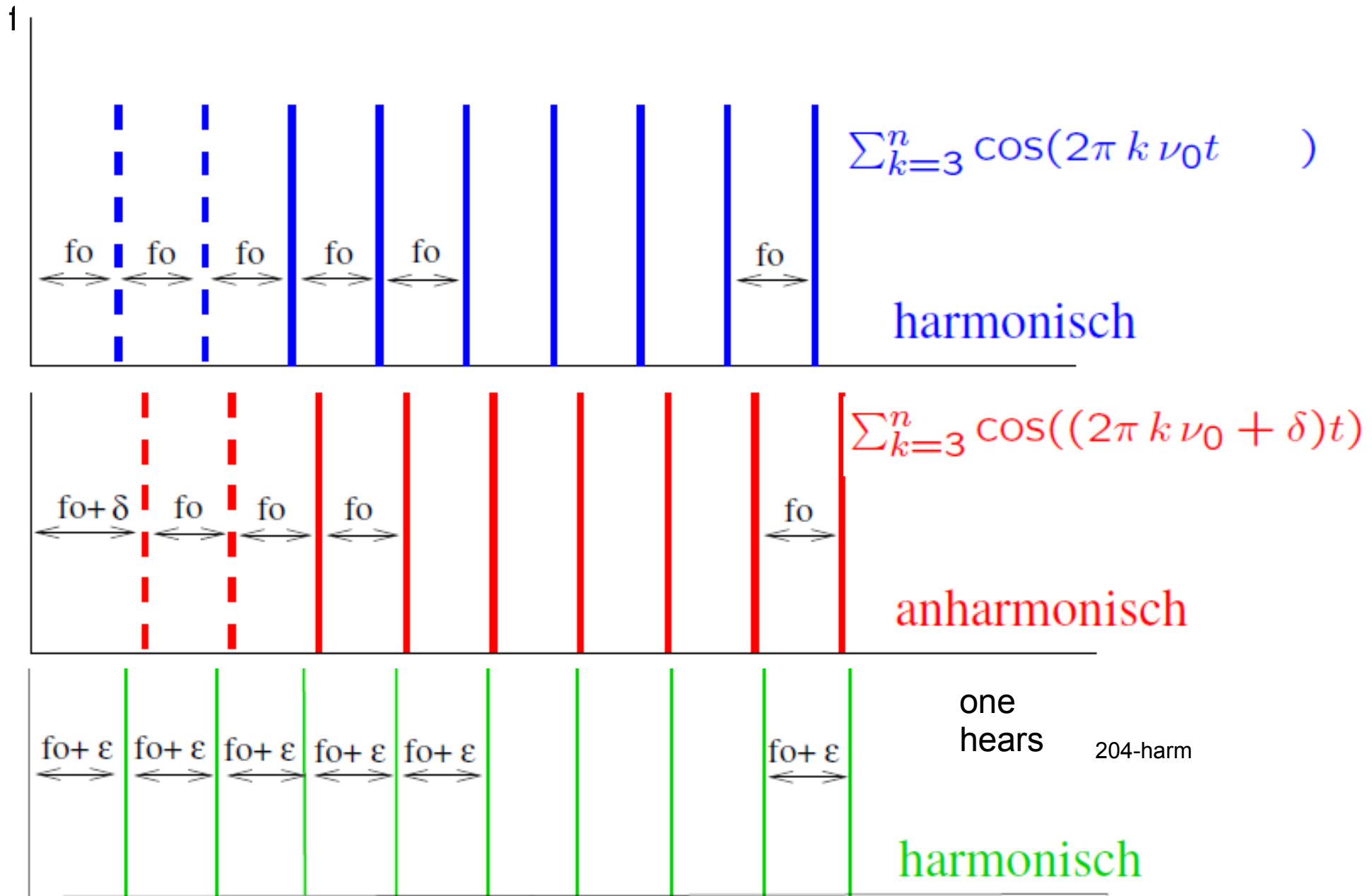
The difference tone was excluded by van Schouten by a series of ingenious experiments. A particular simple one is the shifted harmonic tone.



103praat harm-200 200+40

one  
hears

The difference tone was excluded by van Schouten by a series of ingenious experiments. A particular simple one is the shifted harmonic tone with missing



The fact that the shifted tone has a distinctly different pitch is a sure sign that the difference tone is not responsible for fundamental tracking, since the difference tone is unaffected by the shift

The question what determines the pitch of a complex tone is still controversial. We shall come back to it after looking closer into the physics of the ear and the auditory pathway.

