

Corso di Laurea in Fisica - UNITS
**ISTITUZIONI DI FISICA
PER IL SISTEMA TERRA**

Sound and Fourier analysis

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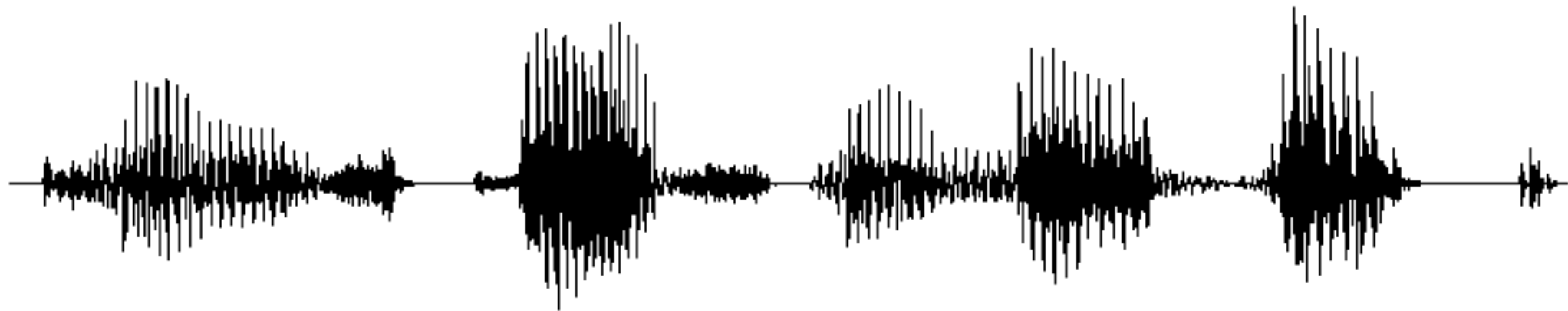
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<http://moodle2.units.it/course/view.php?id=887>



Complex waves

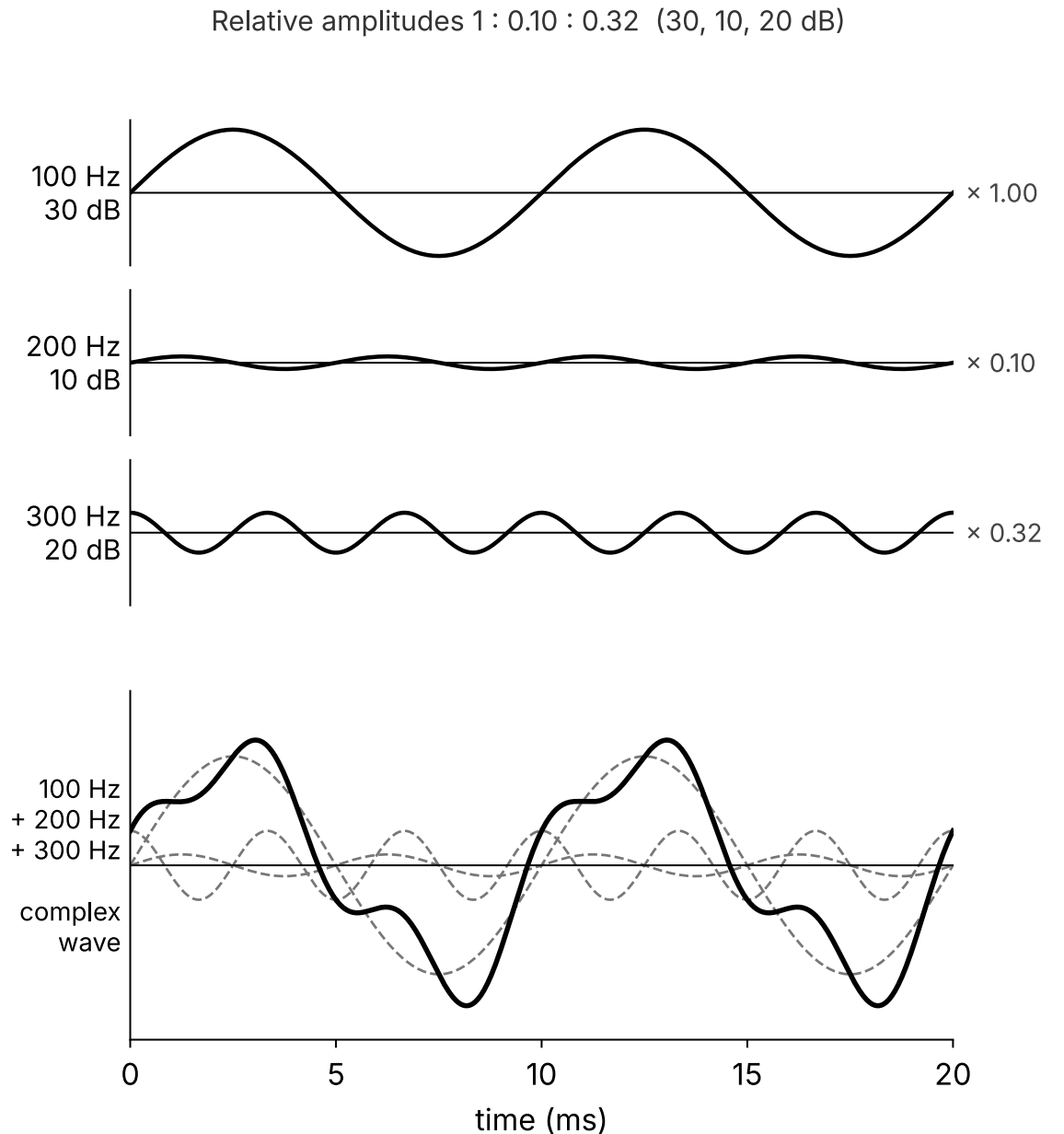
- ☑ Sound waves are not simple sinusoidal waves, but are complex. We can see this visually with a waveform:



Complex waves

Complex waves can be broken down mathematically (by Fourier analysis) into a series of simple waves:

Fourier analysis (simplified): at a given point in time, add the increases in air pressure and subtract the decreases in air pressure





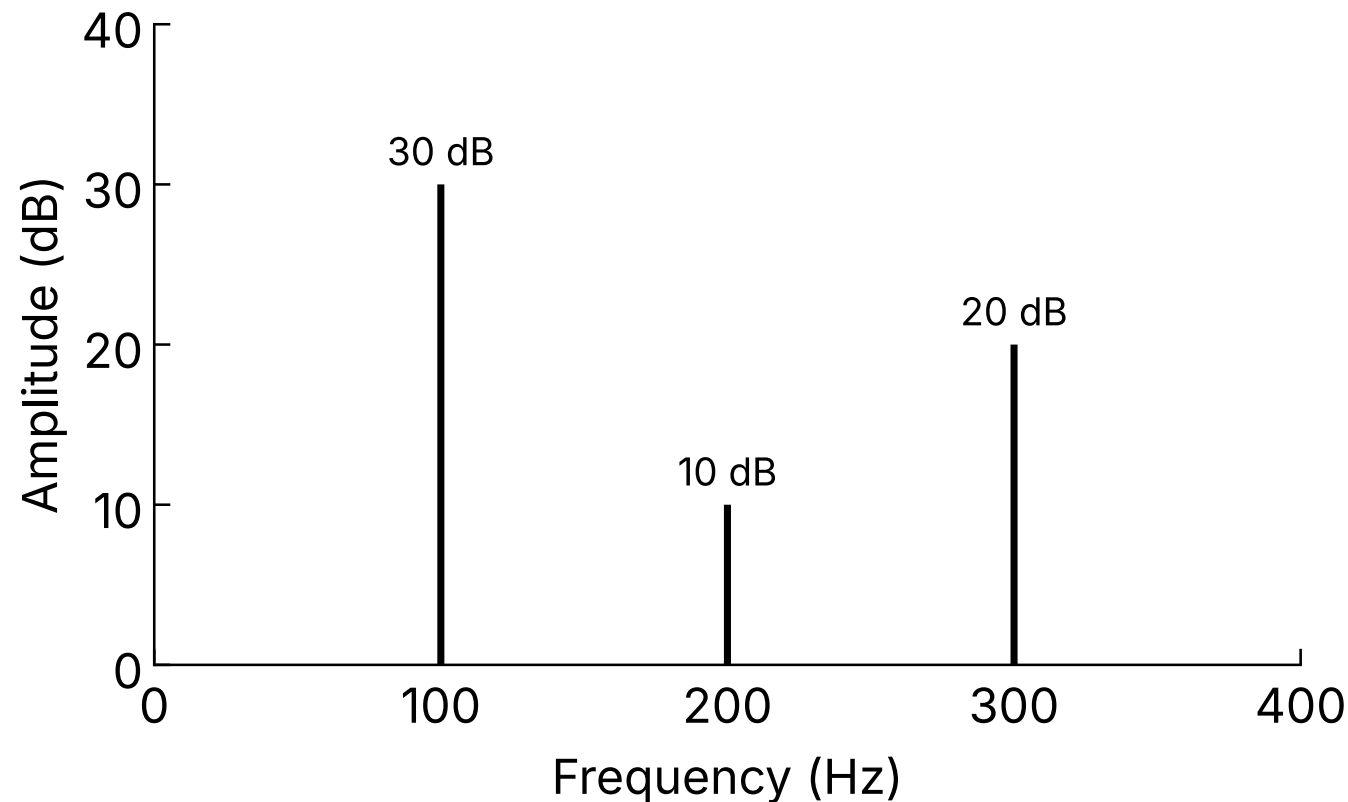
Complex waves



- ☑ The frequency of the complex wave is the fundamental period and is the same as the frequency of the first component of a complex wave. This is the fundamental frequency or f_1 , that corresponds to the “**pitch**” of the sound
- ☑ Each of the higher-frequency simple waves is called a **harmonic**. In naturally occurring vibrations, there is a harmonic at each multiple of the fundamental frequency, so if $f_1 = 100\text{Hz}$,
2nd harmonic = 200Hz, 3rd harmonic = 300Hz , etc...
- ☑ Different patterns of harmonics are part of what contributes to the distinction between the sound of different musical instruments playing the same note

Spectrum

☑ Because each complex wave has many components or harmonics, it is difficult to see in a waveform in which the X-axis is time. We can also plot amplitude with respect to frequency in a **spectrum**, a useful way of interpreting complex waves:



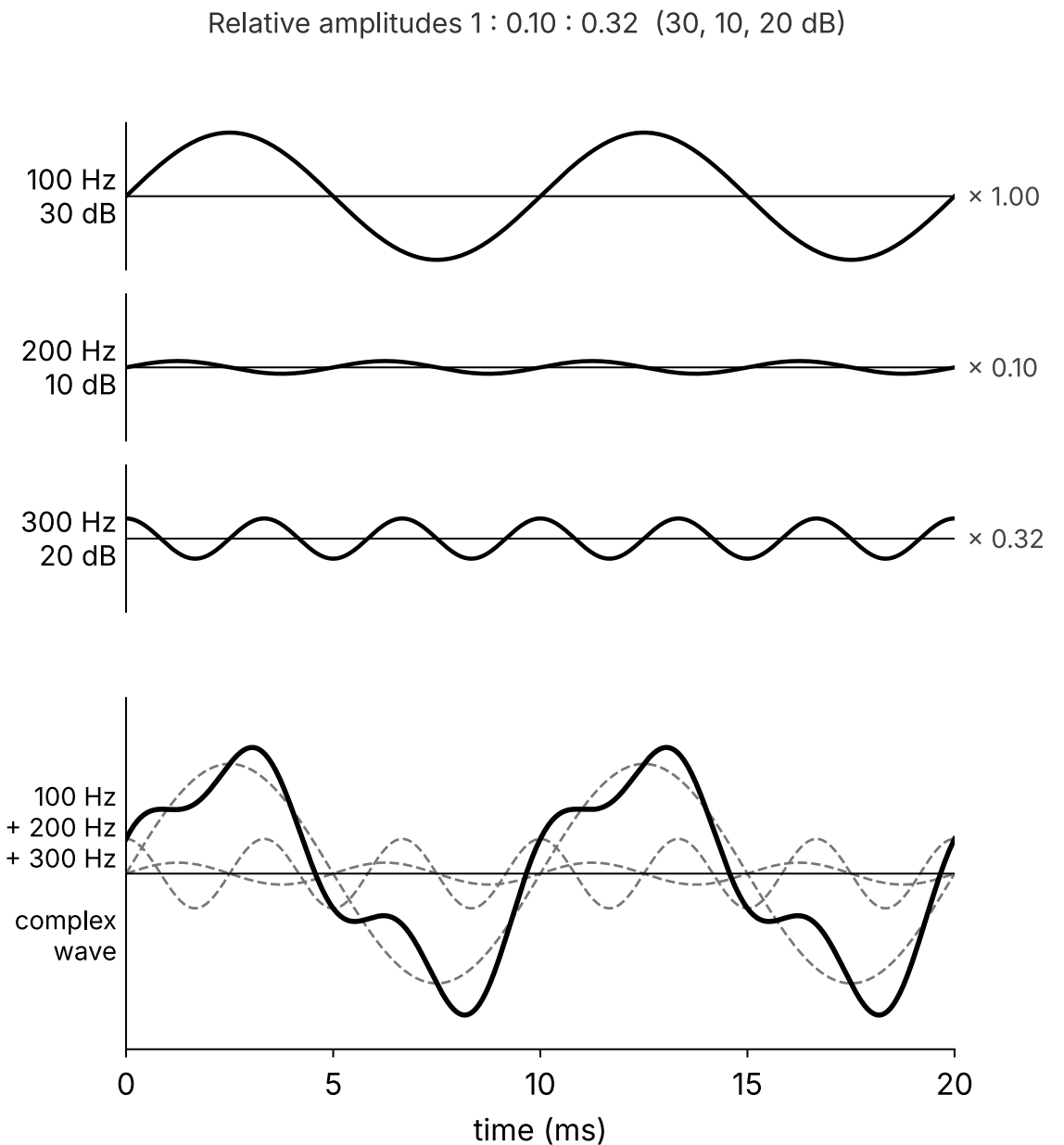
Each line in a spectrum represents a simple wave and indicates its amplitude and frequency

Spectrum

Consider the complex wave again.

It has three components with different frequencies and amplitudes

	Frequency	Amplitude
1	100Hz	30dB
2	200Hz	10dB
3	300Hz	20dB

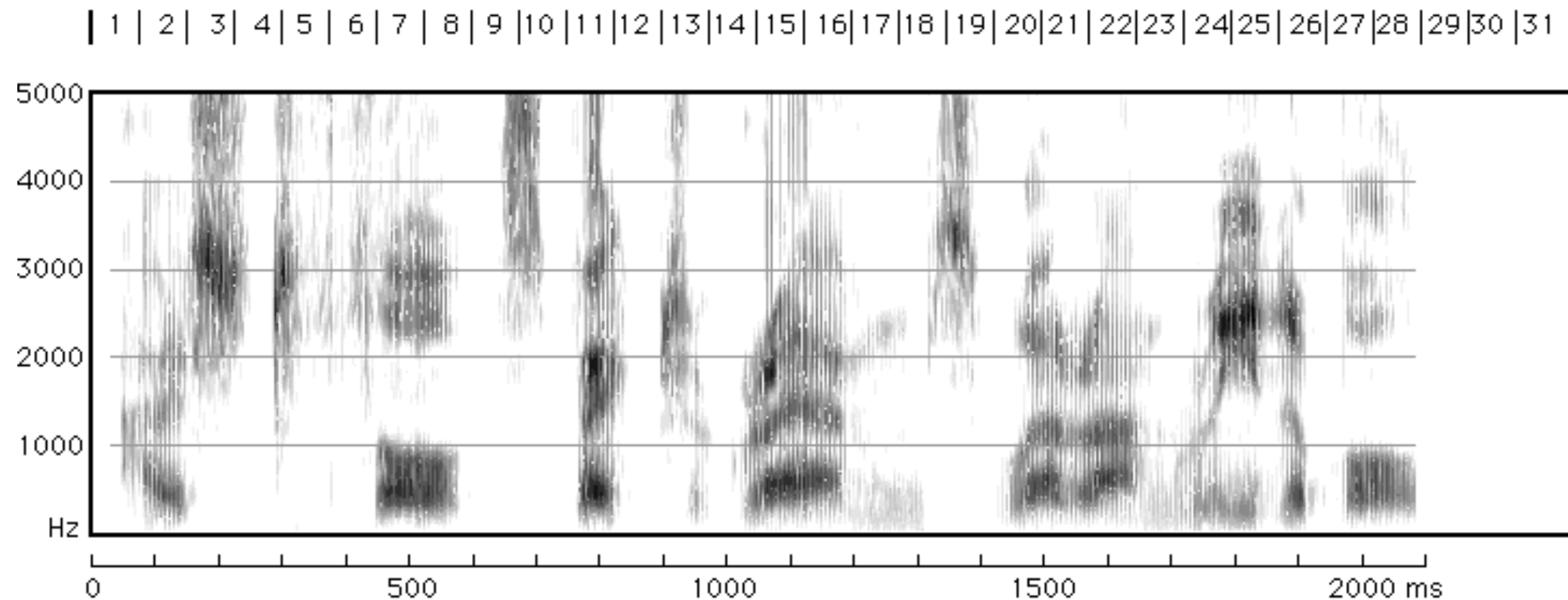


Spectrogram

Another way of examining sound

Spectrogram plots intensity and frequency with respect to time.

X-axis is time; Y-axis is frequency; darkness is intensity:



Fourier Series

✓ Fourier theorem states that a periodic function $f(t)$ which is reasonably continuous may be expressed as the sum of a series of sine or cosine terms (called the Fourier series), each of which has specific amplitude and phase coefficients known as Fourier coefficients.

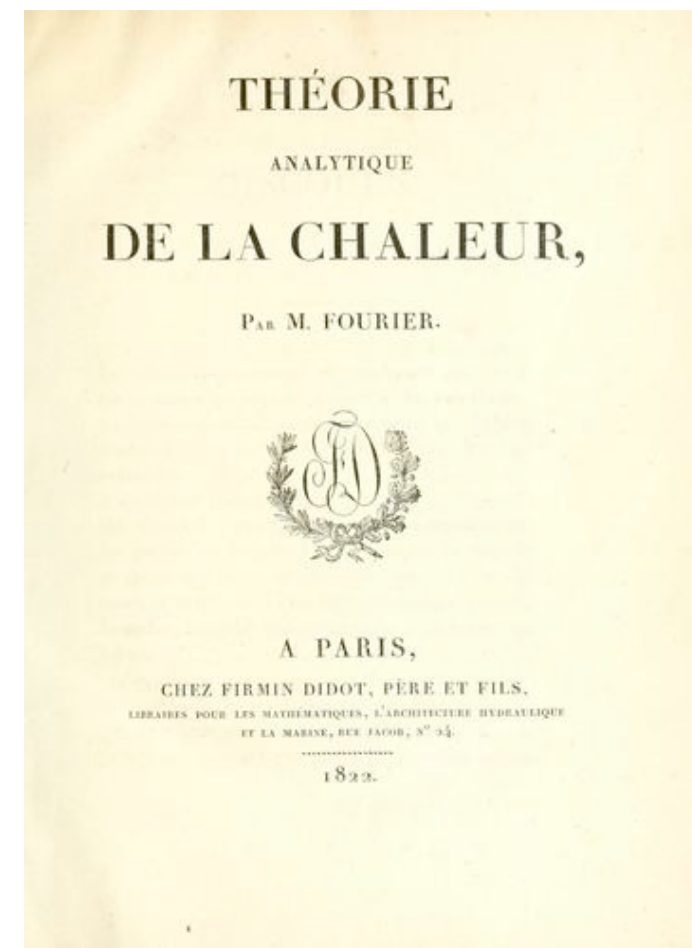
✓ Linear Superimposition of Sinusoids to build complex waveforms

$$f(t) = a_0 + \sum_{n=1}^{\infty} A_n \cos(\omega_n t + \phi_n)$$
$$\omega_n = n\omega_1 = n \frac{2\pi}{T_1}$$

<http://www.falstad.com/fourier/>



Jean Baptiste Joseph Fourier
1768-1830



Fourier Series

✓ Decompose our complex periodic waveform into a series of simple sinusoids

✓ Where

$$f(t) = \frac{1}{2}a_0 + \sum_{n=1}^{\infty} [a_n \cos(n\omega t) + b_n \sin(n\omega t)]$$

$$a_0 = \frac{2}{T} \int_0^T f(t) dt$$

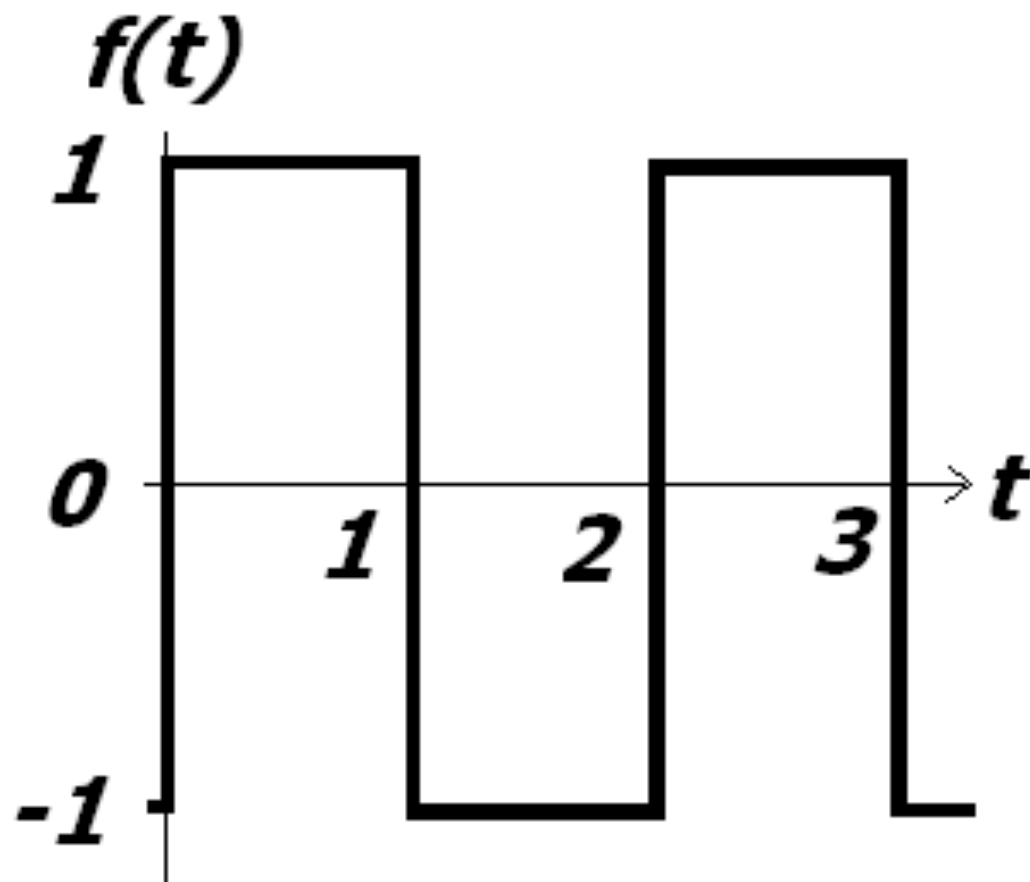
$$a_n = \frac{2}{T} \int_0^T f(t) \cos(n\omega t) dt$$

$$b_n = \frac{2}{T} \int_0^T f(t) \sin(n\omega t) dt$$

$$\omega = \frac{2\pi}{T}$$

Square Wave Example

✓ Consider



$$f(t) = 1 \quad 0 < t < 1$$

$$= -1 \quad 1 < t < 2$$

✓ Clearly the period is $T=2$ hence $\omega=\pi$

✓ When we integrate we need to do so over sections

$t = 0$ to 1 and $t = 1$ to 2


$$a_0 = \int_0^2 f(t) dt$$

• So to find the series, we have to calculate coefficients a_0 , a_n and b_n

$$= \int_0^1 dt - \int_1^2 dt = 1 - 1 \Rightarrow a_0 = 0$$

$$a_n = \int_0^2 f(t) \cos(n\pi t) dt$$

$$= \int_0^1 \cos(n\pi t) dt - \int_1^2 \cos(n\pi t) dt$$

$$= \frac{1}{n\pi} \left[\sin(n\pi t) \Big|_0^1 - \sin(n\pi t) \Big|_1^2 \right] \Rightarrow a_n = 0$$

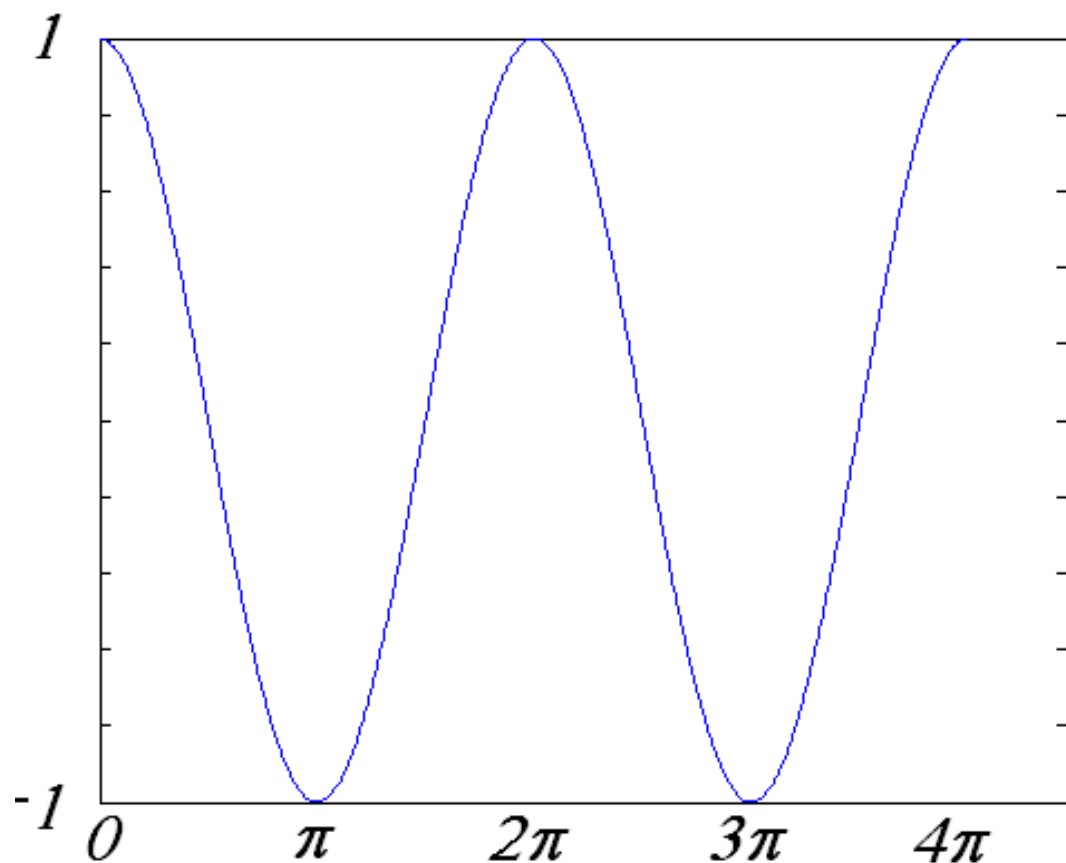
When evaluating a_n note that the sin function is 0 when angle is every multiple of π


$$\begin{aligned} b_n &= \int_0^2 f(t) \sin(n\pi t) dt = \int_0^1 \sin(n\pi t) dt - \int_1^2 \sin(n\pi t) dt \\ &= -\frac{1}{n\pi} \left[\cos(n\pi t) \Big|_0^1 - \cos(n\pi t) \Big|_1^2 \right] \\ &= -\frac{1}{n\pi} \left((\cos(n\pi) - 1) - (\cos(n\pi 2) - \cos(n\pi)) \right) \\ &= -\frac{1}{n\pi} (2 \cos(n\pi) - 1 - \cos(n2\pi)) \end{aligned}$$

✓ Knowing

$$b_n = -\frac{1}{n\pi} (2 \cos(n\pi) - 1 - \cos(n2\pi))$$

✓ We need to consider the cos function to determine values of b_n for $n=1,2,3,\dots$ etc



$$n = 1, \quad b_1 = -\frac{1}{n\pi} (2(-1) - 1 - 1) = \frac{4}{n\pi}$$

$$n = 2, \quad b_2 = -\frac{1}{n\pi} (2(1) - 1 - 1) = 0$$

$$n = 3, \quad b_3 = -\frac{1}{n\pi} (2(-1) - 1 - 1) = \frac{4}{n\pi}$$

$$n = 4, \quad b_4 = -\frac{1}{n\pi} (2(1) - 1 - 1) = 0$$

$$n = 5, \quad b_5 = -\frac{1}{n\pi} (2(-1) - 1 - 1) = \frac{4}{n\pi}$$



☑ We found coefficients to be

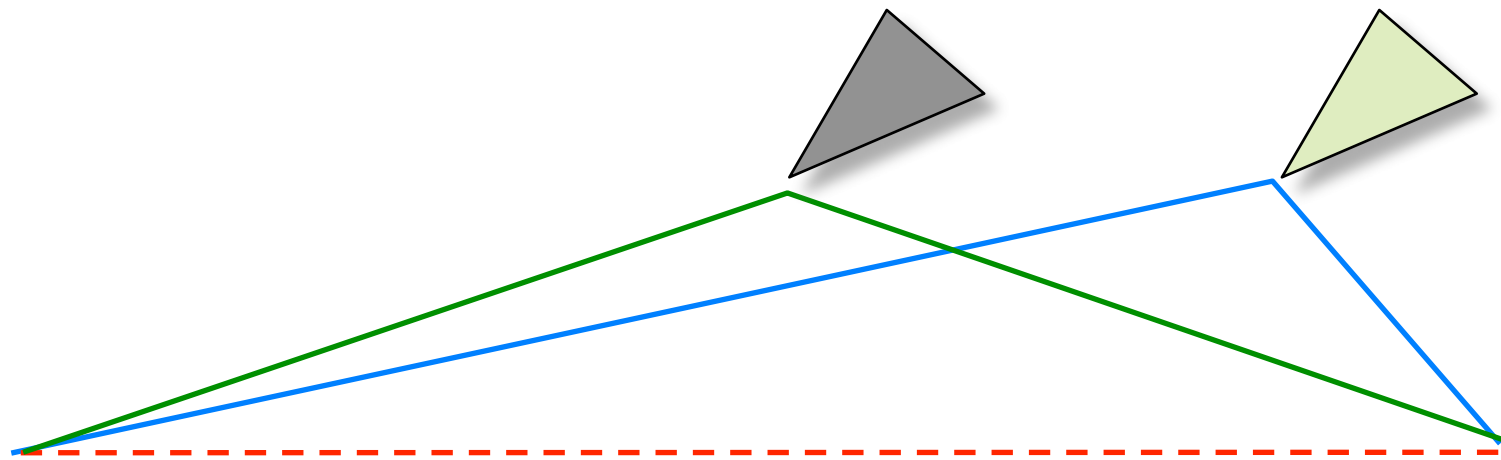
$$a_0 = 0 \quad a_n = 0 \quad b_n = \frac{4}{n\pi} \text{ when } n = 1, 3, 5, \dots$$

☑ Hence Fourier Series for a square wave is

$$\begin{aligned} f(t) &= \frac{1}{2} a_0 + \sum_{n=1}^{\infty} [a_n \cos(n\omega t) + b_n \sin(n\omega t)] \\ &= \frac{4}{\pi} \left[\sin(\pi t) + \frac{1}{3} \sin(3\pi t) + \frac{1}{5} \sin(5\pi t) + \dots \right] \end{aligned}$$

Plucked string

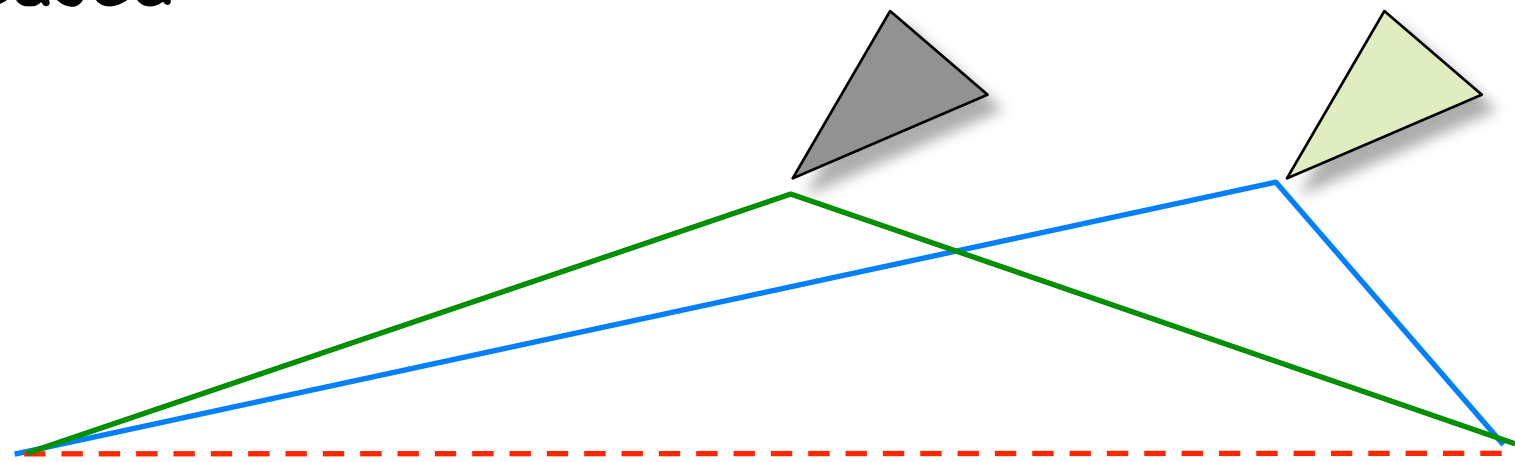
- ✓ Can one predict the amplitude of each mode (overtone/harmonic?) following plucking?



- ✓ Using the procedure to measure the Fourier coefficients it is possible to predict the amplitude of each harmonic tone.

Plucked string

- You know the shape just before it is plucked.
- You know that each mode moves at its own frequency
- The shape when released
- We rewrite this as



$$f(x, t = 0)$$

$$f(x, t = 0) = \sum_{m=0}^{\infty} A_m \sin \frac{2\pi m x}{2L}$$

Plucked string

Each harmonic has its own frequency of oscillation, the m -th harmonic moves at a frequency $f_m = mf_0$ or m times that of the fundamental mode.

$$f(x, t = 0) = \sum_{m=0}^{\infty} A_m \sin \frac{2\pi mx}{2L} \quad \text{initial condition}$$

$$f(x, t) = \sum_{m=0}^{\infty} A_m \sin \frac{2\pi mx}{2L} \cos 2\pi mf_0 t$$

<http://www.falstad.com/loadestring/>

Modal summation on a string

Recall modes on a string:

$$u(x, t) = \sum_{n=0}^{\infty} A_n U_n(x, \omega_n) \cos(\omega_n t)$$

This is the sum of standing waves or *eigenfunctions*, $U_n(x, \omega_n)$, each of which is weighted by the amplitude A_n and vibrates at its *eigenfrequency* ω_n .

The eigenfunctions and eigenfrequencies are constants due to the physical properties of the string.

The amplitudes depend on the position and nature of the source that excited the motion.

The eigenfunctions were constrained by the boundary conditions, so that

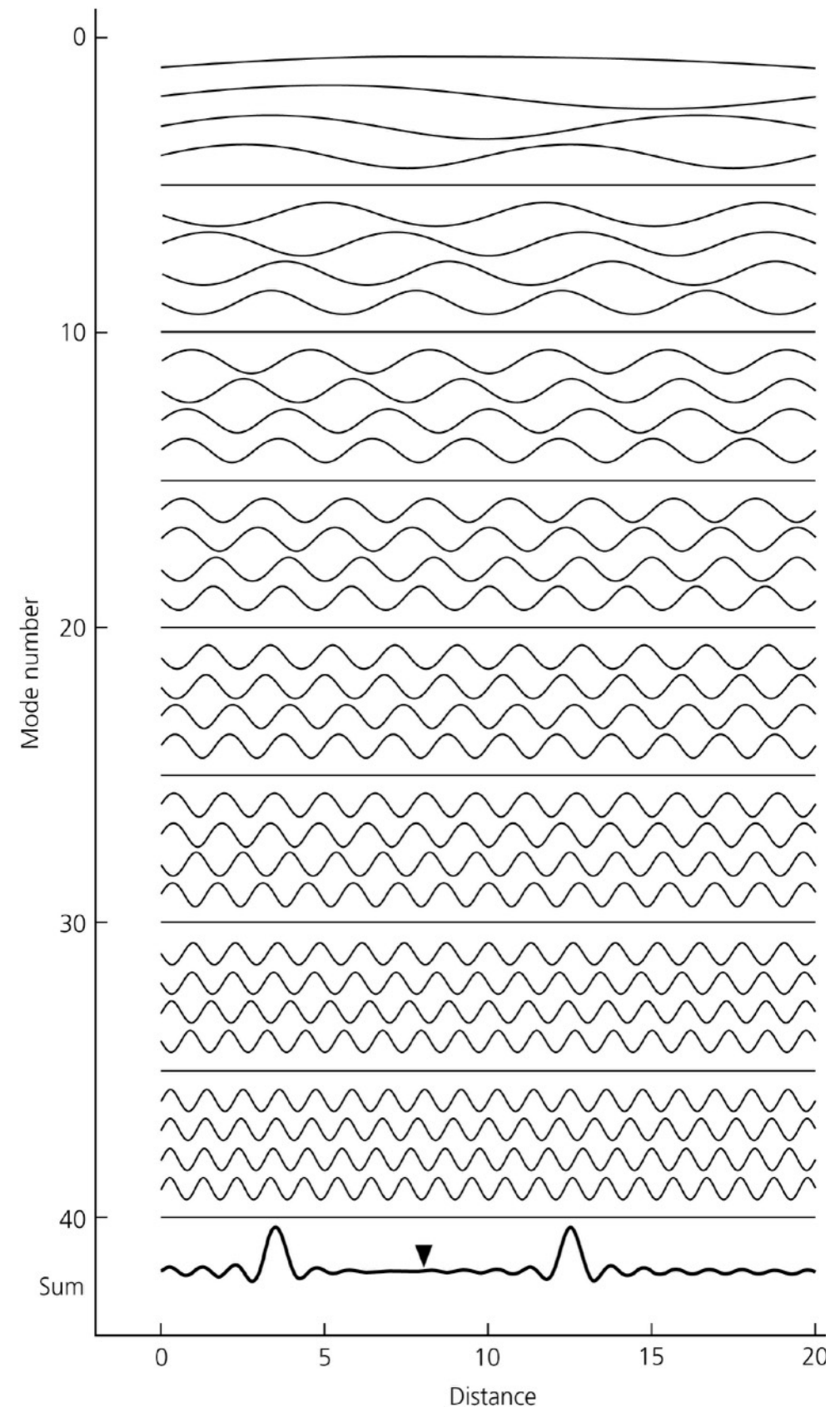
$$U_n(x, \omega_n) = \sin(n\pi x/L) = \sin(\omega_n x/v) \qquad \omega_n = n\pi v/L = 2\pi v/\lambda$$

$$u(x, t) = \sum_{n=0}^{\infty} \sin(n\pi x_s/L) F(\omega_n) \sin(n\pi x/L) \cos(\omega_n t)$$

The source, at $x_s = 8$, is described by

$$F(\omega_n) = \exp[-(\omega_n \tau)^2/4]$$

with $\tau = 0.2$.





Do we hear phase?



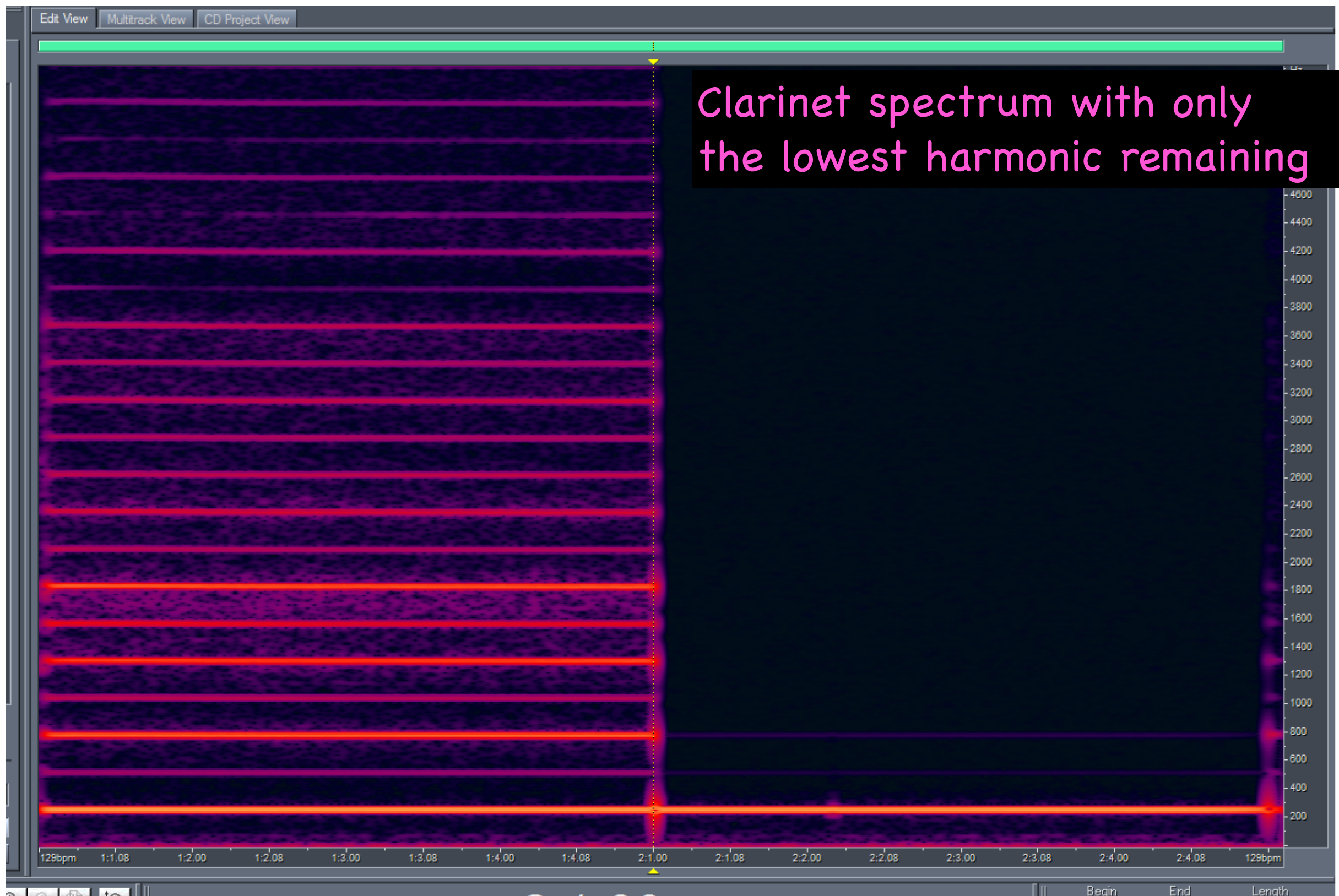
Helmholtz and Ohm argued that our ear and brain are only sensitive to the frequencies of sounds. Timbre is a result of frequency mix.

These two are sums with the same amplitude sine waves components, however the phases of the sine waves differ.

This sound file has varying phases of its frequencies. Do we hear any difference in time?

Clarinet spectrum

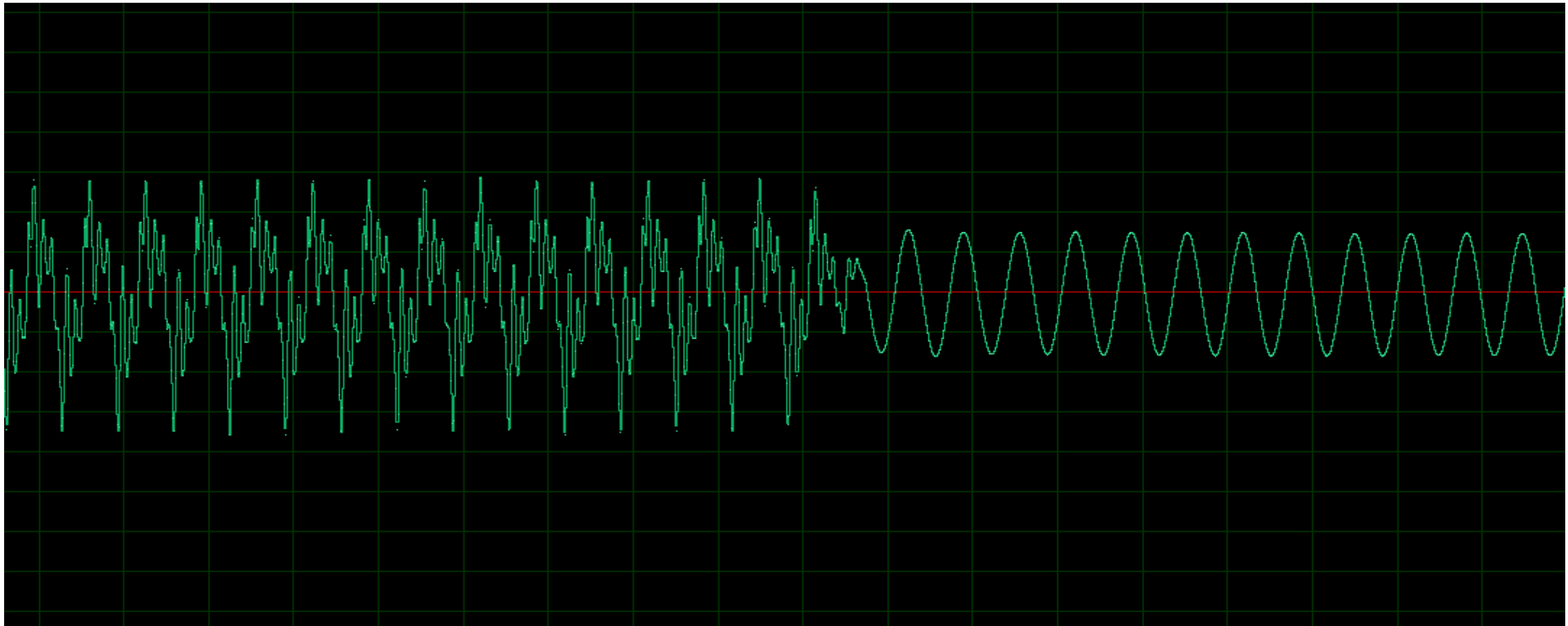
Spectral view



Waveform view

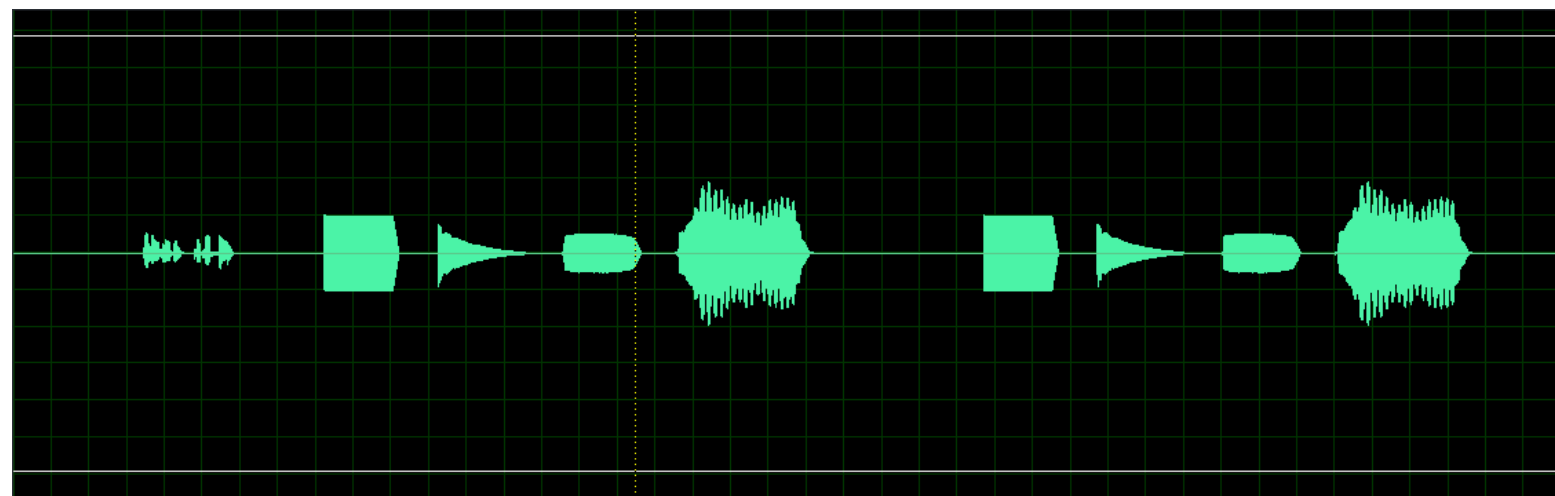
Full sound

Only lowest harmonic



Four complex tones in which all partials have been removed by filtering

One is a French horn, one is a violin, one is a pure sine, one is a piano (but out of order)



It's hard to identify the instruments. However clues remain (attack, vibrato, decay)