

AUD401

Dynamic Audio for Digital Media

Lecture 4 ~ Sound Synthesis

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So far:

1. Counters
2. Digital Audio
3. Loading Samples
4. Playing them Back

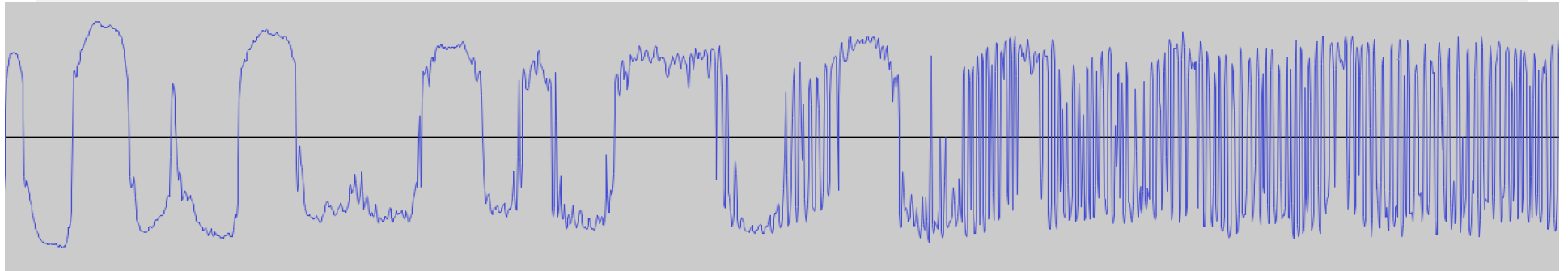
Today

Additive Synthesis
FM Synthesis

But first

Sound Synthesis Fundamentals

Pitch / Noise



Some definitions

Fundamental

The lowest frequency of a periodic waveform

Overtones

All sinusoidal peaks above that

Partials

All of the above

Pitched:

Harmonic / Inharmonic Sound

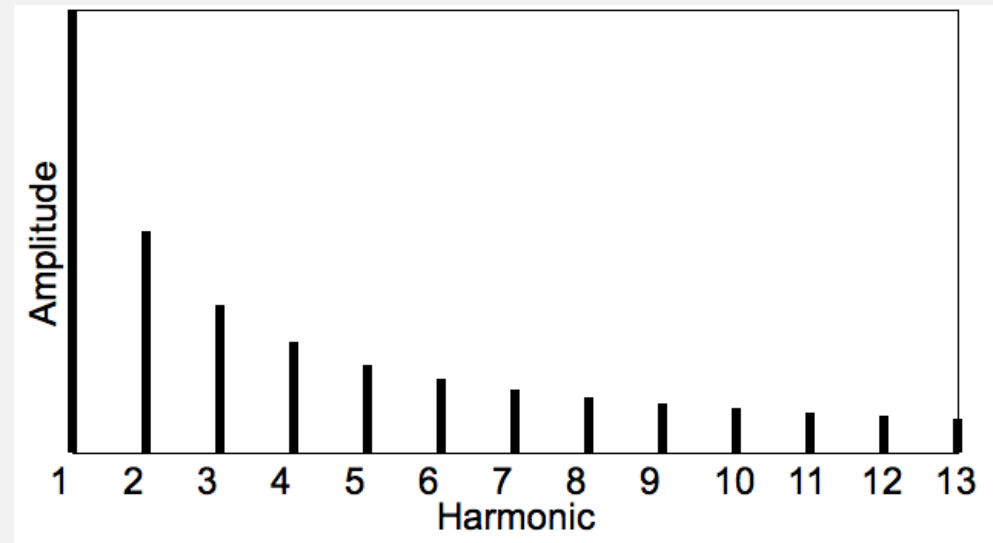
Harmonic

- Clear Sinusoidal peaks
- Integer multiples

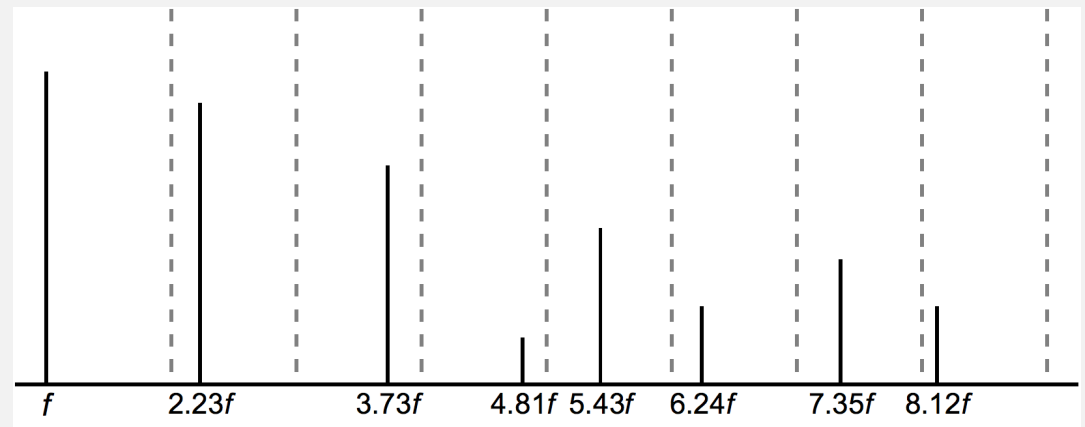
Inharmonic

- Clear Sinusoidal peaks
- **NOT** integer multiples

Harmonic



Inharmonic



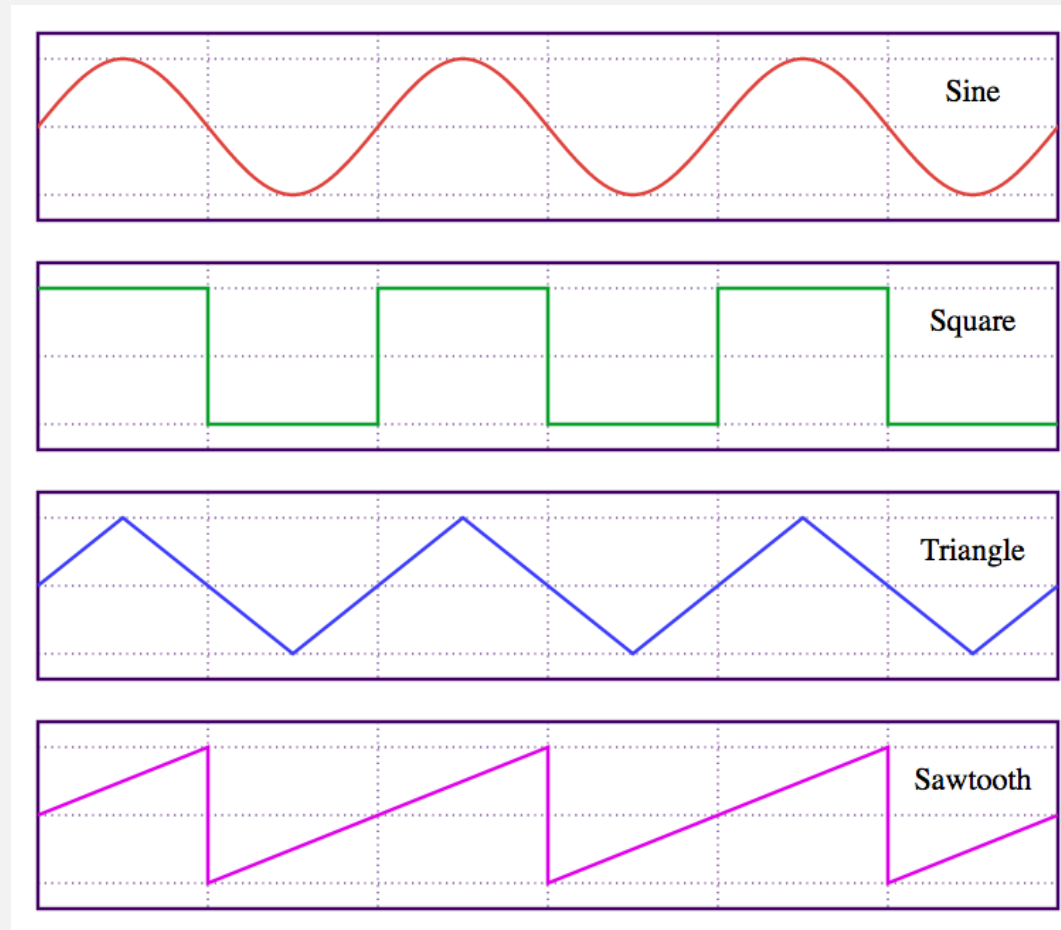
Fourier Series

Any periodic waveform can be represented as a sum of simple sine waves

Mathematician **Joseph Fourier** 1768-1830



How can we express our basic waveforms by adding sinusoids?



...we can do this using **Wavetables**

Wavetables

- Storage of waveform data in an array (i.e. memory)
- Stored as amplitude vs time: *i.e.* indexed by time
- Lookup data at a particular frequency (sample the wavetable)
- Use interpolation to produce values not directly represented in the table

Wavetables

- Useful for static waveforms as often takes less computation to lookup a table value, than directly calculate values from waveform functions
- Can crossfade between several wavetable oscillators to create varying timbres

Wavetables

- In pd, create an **array** to hold your wavetable in
- You will need to decide on a length. If you are using **tabosc4~**, the length of the array should be a power of two plus 3 extra points to allow for the 4-point interpolation.
e.g. 131, 259, 515, 1027
- You can draw the waveform directly into the table

Wavetables

Array messages:

sinesum creates the weighted sum of sine waves and puts the results into the array. It takes the form:

sinesum length a(0) a(1) a(2) ... a(x)

Where **length** is the length of the array, and **a(0)** the amplitude of the fundamental, **a(1)** the amplitude of the first harmonic, *etc.* **NB** unlike cosinesum there is **no** offset argument!

Wavetables

With **sinesum**:

- Notice that it **resizes** your array to fit the length specified. It automatically add the guard points required for interpolation!
- Watch that the total magnitude does not go above 1.0 or you will get distortion. You can send a `normalize` message to the table to get the waveform within the proper bound

Wavetables

Other messages:

- **const** set array to constant value
- **normalize** normalize array to certain value

tabosc4~

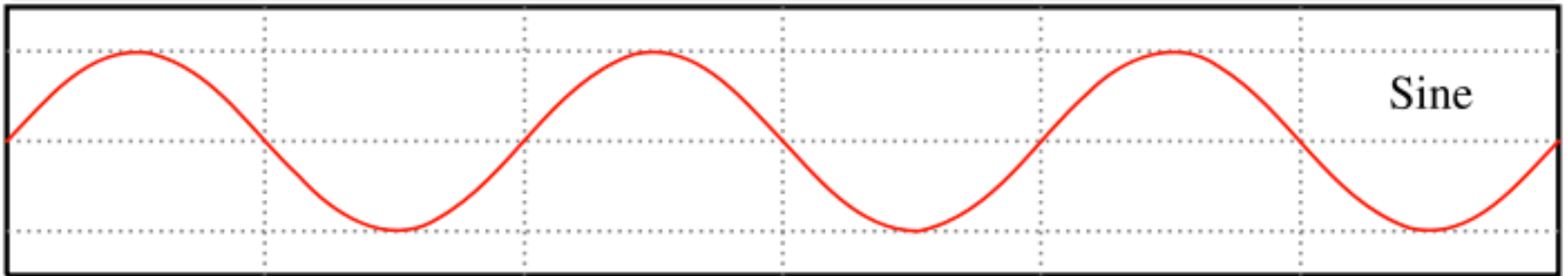
- Leftmost inlet: frequency
- Right inlet: reset phase (*cf* osc~)
- Use set messages to switch between tables (do not use sinesum or cosinesum messages to array while running to change timbre!)

Again:

Any periodic waveform can be represented
as a sum of simple sine waves

...so let's start adding sine waves

Sine Wave



- Fundamental only
- No additional harmonics

Sawtooth Wave



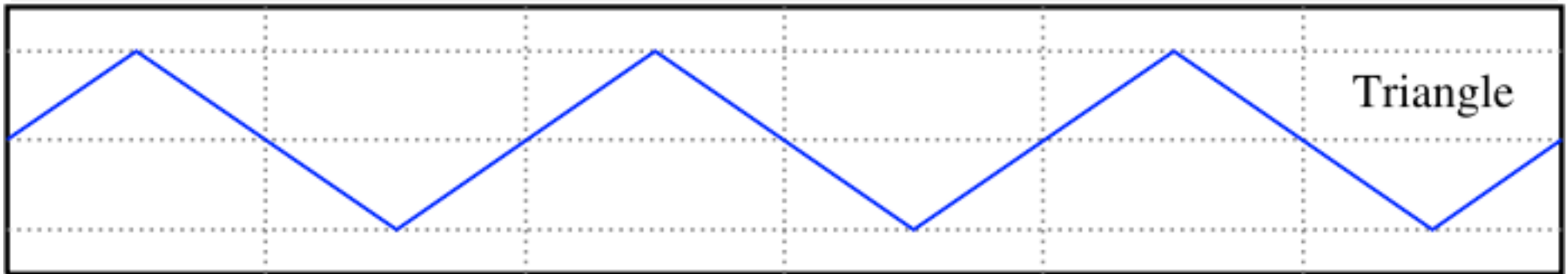
- All partials
- Partial amplitude = $1/p\#$

Square Wave



- Odd-numbered partials only
- Partial amplitude = $1/p\#$

Triangle Wave



- Odd-numbered partials only
- Partial amplitude = $1/p\#^2$
- Amplitude of every other harmonic multiplied by -1

Additive Synthesis!

Additive Synthesis

- The basic force behind the use of additive synthesis is a desire to create complex musical sounds by adding (mixing) together multiple, simpler, sound components.
- The component sounds are not usually perceived, but instead contribute to the quality (timbre) of the synthesised resulting sound.

Additive Synthesis

- The theoretical framework for additive synthesis based on Fourier analysis.
- To put Fourier's theorem in basic terms: periodic (acoustic) waveforms can be expressed as a sum of harmonically related sine waves, each with a particular phase and amplitude. This sum may, however, be infinite!

Additive Synthesis

- Although sine waves are often used as the building blocks for additive synthesis, different waveforms may also be used to create more complex results.

Frequency Modulation Synthesis (FM)

FM Synthesis

- The 'carrier' frequency is being frequency modulated.
- Terms:
 - Carrier** — waveform being modulated.
 - Modulation Frequency**—rate of modulation.
 - Modulation Index**—how much the signal is modulated by.

FM Synthesis

- Discovered by John Chowning in the early 70's and patented to Yamaha (used in, for example, the Yamaha DX7).
- In its simplest form, it comprises a sine wave carrier whose frequency is varied (modulated) by another waveform (e.g. another sine wave).
- When the modulation frequency is sub-audio (below c. 20Hz, the change in pitch is perceptible.

FM Synthesis

If **f_m / f_c** is a positive integer, then the spectrum of the resulting waveform will be **harmonic**.

- If **f_m / f_c** is a positive non-integer, then the spectrum will be **inharmonic**. This can be useful for synthesizing things such as bells, gongs and even drums.

Tutorial