### **DSP for Digital Artists**

a 3-Day Short Course, by Perry R. Cook, PhD CalArts Intersession January 14-18, 2019

Day 2: More Filters, Feedback Filters, More Sine Waves, Spectrum Analysis, Modal Synthesis

# **Some More Time-Domain "Pitch" Detection:**

**AMDF:** pick m that minimizes: **Autocorrelation:** pick m that maximizes:

 $y(n) = \sum_{n} |x(n) - x(n-m)|$   $y(n) = \sum_{n} x(n) * x(n-m)$ 

## Impulse Response:

#### Transfer Function:

h(n) = y(n), in response to input  $x(n) = \delta(n)$  H(f) = Y(f), for every  $x(t) = \sin(2\pi ft)$ 

 $\delta$  (n) = 1 if n==0, 0 otherwise Digital: H(n) = Y(n), for x(n) =  $\sin(2\pi f n/SR)$ 

# **Linear, Time Invariant (LTI) Systems**

Linearity: Homogeneity ( $\alpha x \rightarrow \alpha y$ ) and Superposition ( $x_1 + x_2 \rightarrow y_1 + y_2$ )

Time Invariance:  $x(\tau + t) \rightarrow y(\tau + t)$  for all  $\tau$ 

OnePole:  $(1^{st} \text{ Order feedback}) \quad y(n) = gx(n) + ry(n-1) \quad \text{NOTE: } r \le 1.0 \quad !!!$ 

Set r and g exactly 1.0 yields "Digital Integrator" Set 0.0 < r < 1.0 and g = (1.0 - r) Low-pass filter High-pass Cousin: set -1.0 < r < 0.0 High-pass filter

**Convolution**  $y(n) = \sum_{m} x(n-m) h(m)$  implement Impulse Response as FIR filter

**Filters:** y(n) = F(x(n)) Operations on Digital Signals

Popular Simple Filters: Linear Gain: y(n) = g x(n) where g = some constant

**Moving Average**: y(n) = 0.5x(n) + 0.5x(n-1) (low-pass) High-Pass Cousin: y(n) = x(n) - x(n-1) (digital differentiator)

Also called **OneZero Filter**, Averaging more samples: More Zeroes!!

Also type of Finite Impulse Response (FIR) Filter

Recursive Filters: Feedback, Auto Regressive, IIR, Pole(s)

OnePole Filter: y(n) = g x(n) + r y(n-1) NOTE: | r | < 1.0

NPole Filter:  $y(n) = g x(n) + \sum_{m=1 \text{toN}} -b_m y(n-1)$ 

**Super Special Filter, 2-Pole "Resonator"**  $y(n) = g x(n) + 2r cos(2\pi f/SR) y(n-1) - r^2 y(n-2)$ 

**Modal Synthesis:** Modes = natural resonances of system. Excite those and let them ring!

Code and Demos: See Day2.zip

<u>Assignment:</u> Impulse Responses & Modes: Look around, listen, find some systems. Find/Record Modal sound. Use FFTFindModes to analyze it. If you're brave, use FFTResynth (Noise and/or Residue) to recreate it.

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