



Lecture 6: Linear Prediction (LPC)

1. Resonance & The Source-Filter Model
2. Linear Prediction (LPC)
3. LP Representations
4. LP Synthesis & Modification

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I. Resonance

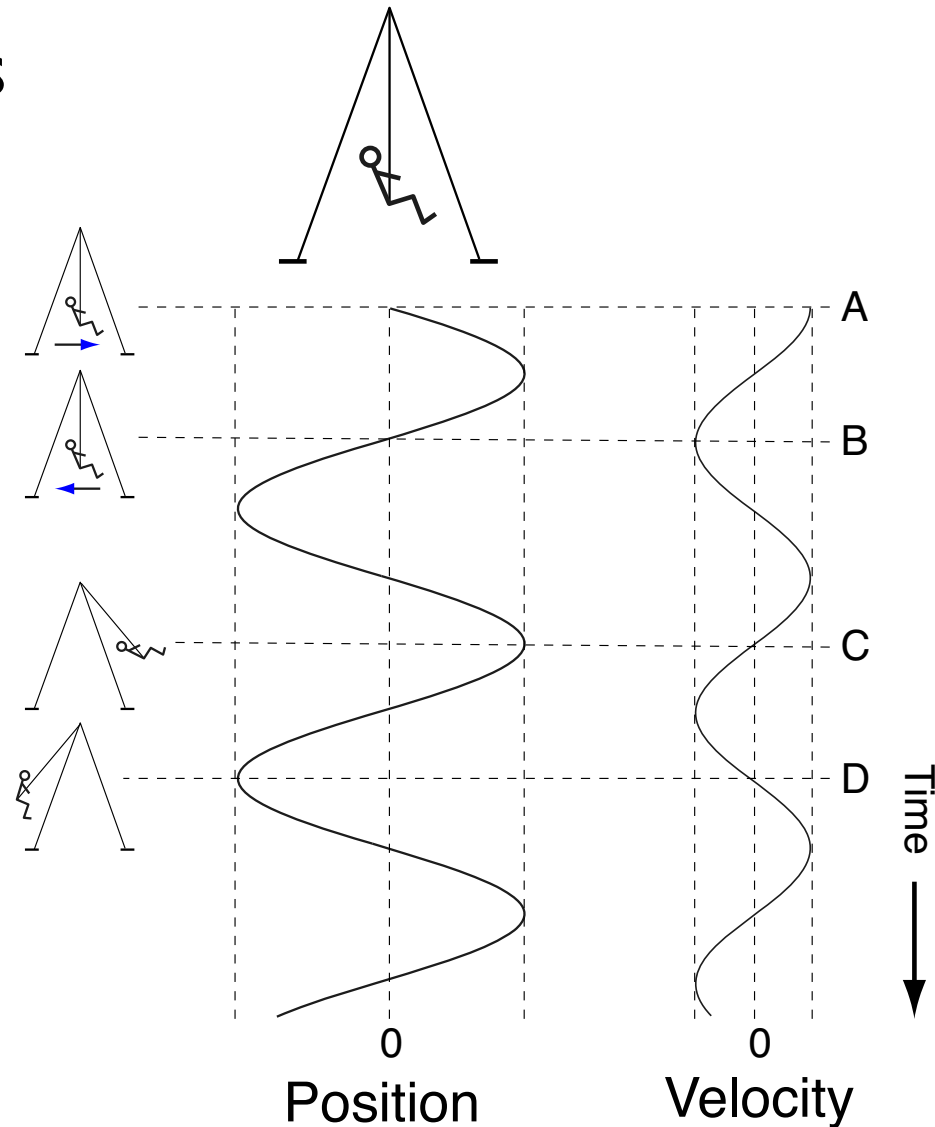
- **Resonance** is ubiquitous in physical systems

- e.g. plucked/struck string, drum head
- room “coloration”
- vocal tract

- **Resonances**

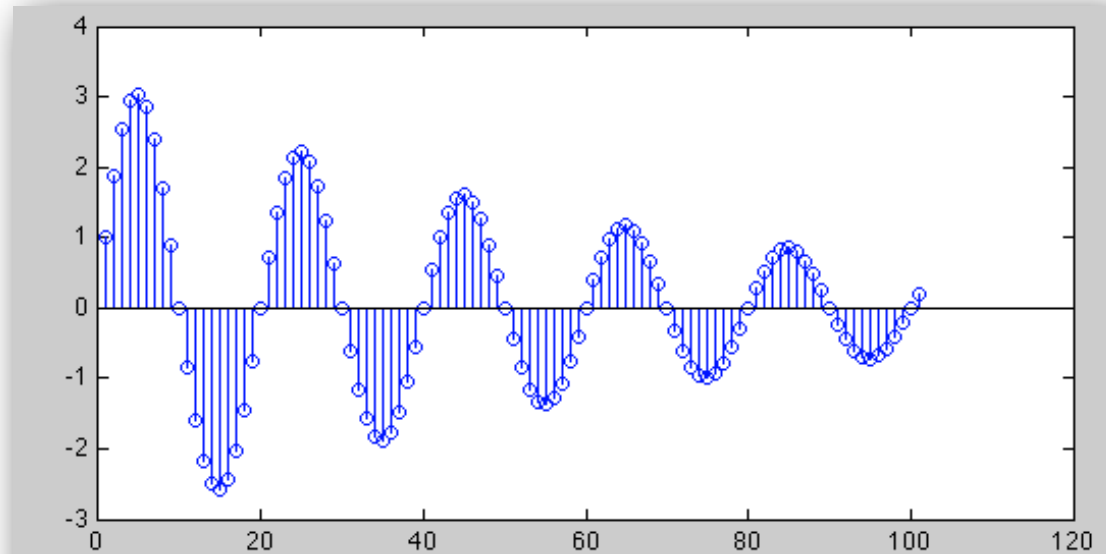
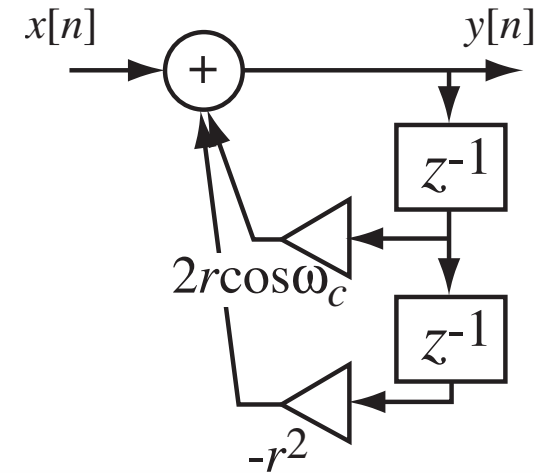
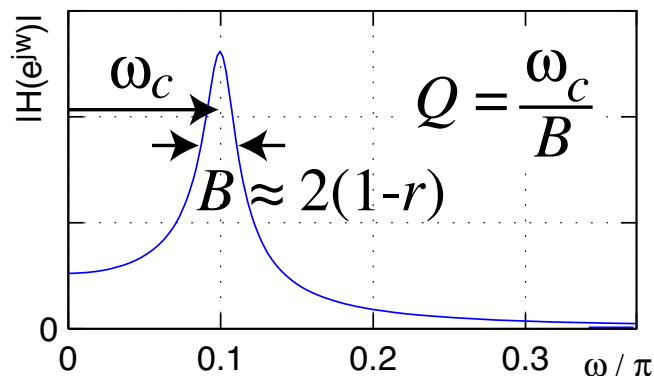
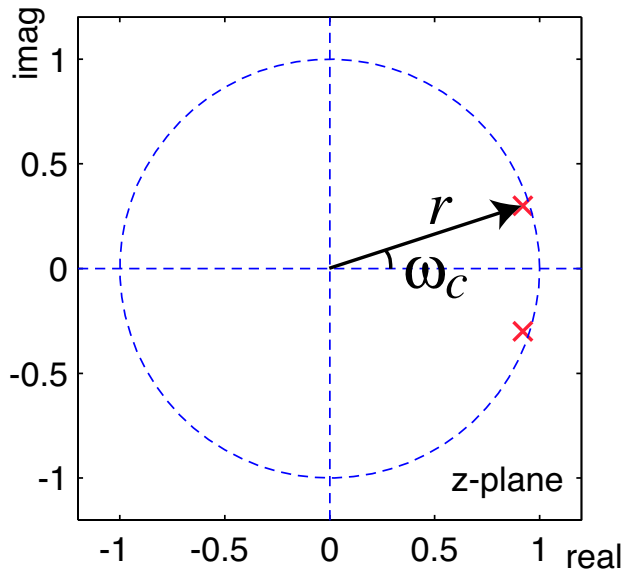
↔ **Poles**

- easily implemented in LTI filters



Singe Pole-Pair Resonance

- Simple **resonance** =
Second-order **IIR**
Band Pass Filter



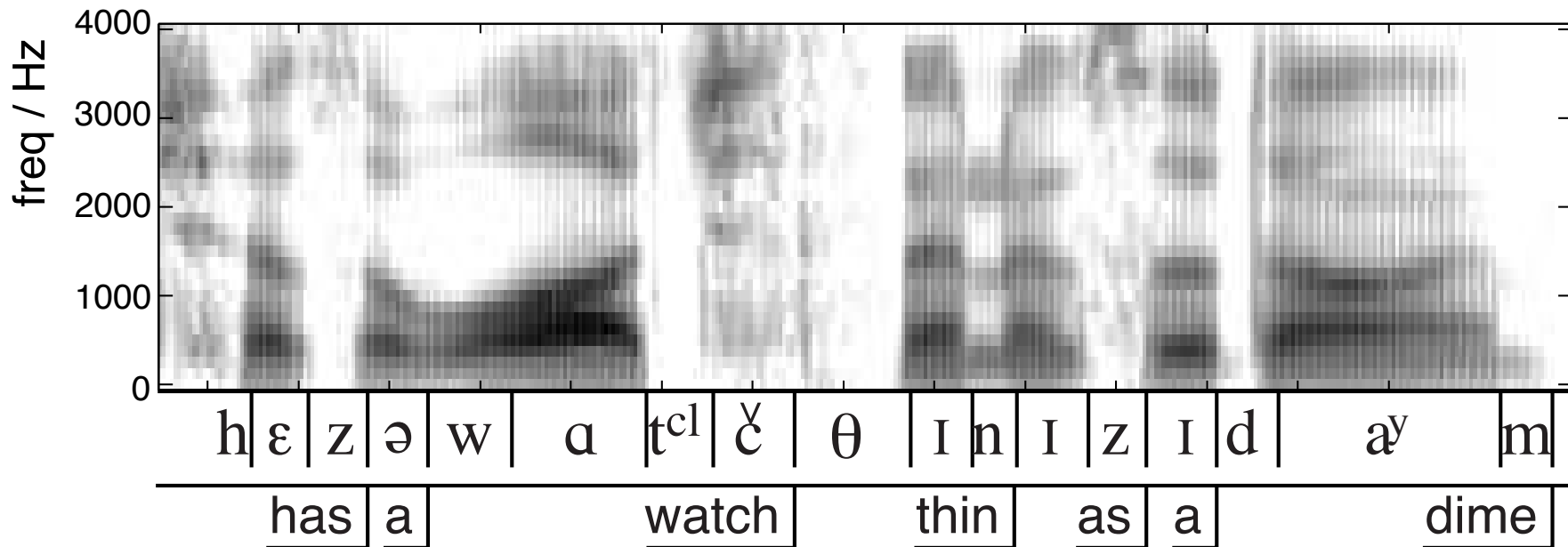
```
>> Q = 10;
>> omega = pi*0.1;
>> r = 1 - omega/(2*Q);
>> a = [1 -2*r*cos(omega) r*r];
>> y = filter(1,a,[1,zeros(1,100)]);
>> stem(y)
```



impulse_bpf.pd

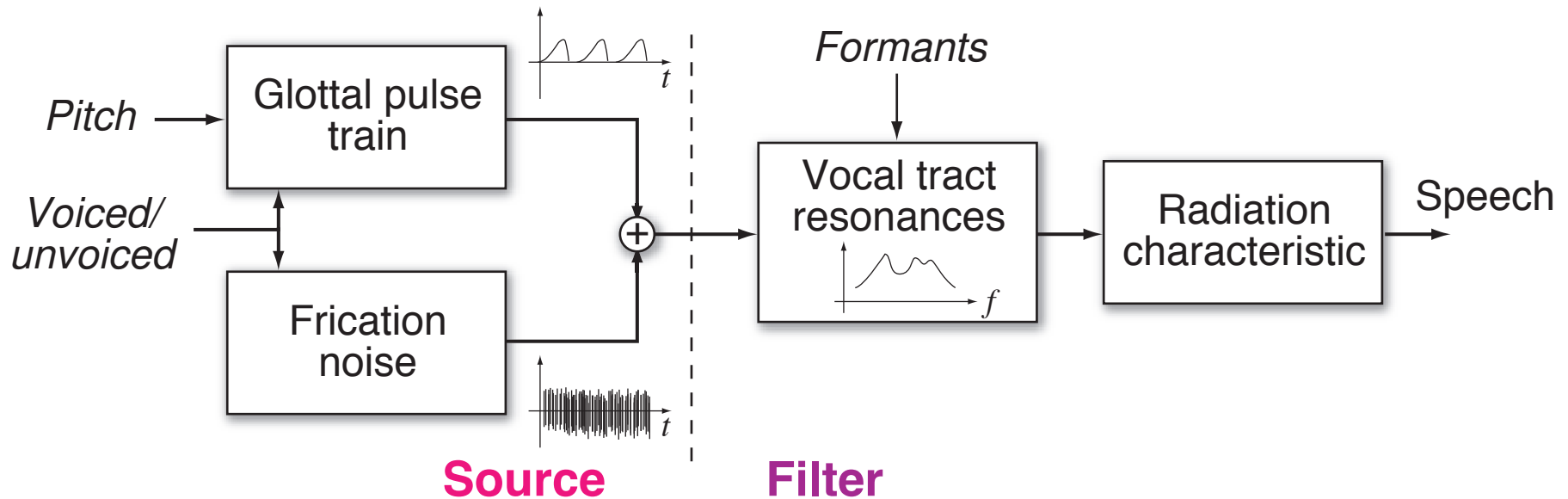
Resonances in Speech

- Vocal Tract (throat + tongue + lips) acts as **variable resonator**
 - resonances = “**formants**”



Source-Filter Model

- Separation of:
 - Source: fine structure in time/frequency
 - Filter: subsequent shaping by physical resonances



- **Advantages**
 - Good match to real signals
 - Salient pieces

2. Linear Prediction (LPC)

- **LPC = Linear Predictive Coding**
 - remove **redundancy** in signal
 - try to predict next point as **linear combination** of **previous values**

$$s[n] = \sum_{k=1}^p a_k s[n-k] + e[n]$$

- $\{a_k\}$ are p^{th} order **linear predictor** coefficients
 - $e[n]$ is residual “innovation” a/k/a **prediction error**
- **Transfer function**

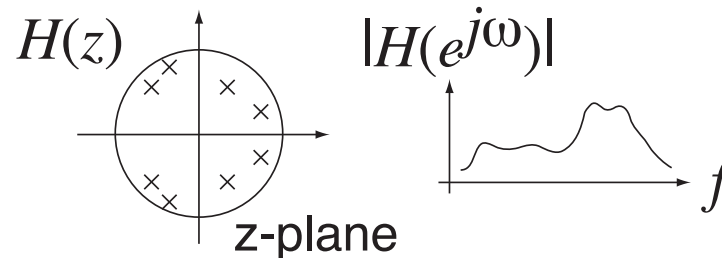
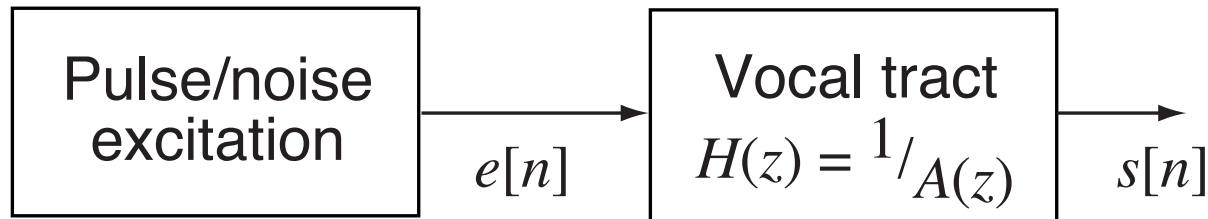
$$\frac{S(z)}{E(z)} = \frac{1}{1 - \sum_{k=1}^p a_k z^{-k}} = \frac{1}{A(z)}$$

- all-pole “autoregressive” (AR) modeling

Voice Modeling & LPC

- Direct expression of **source-filter** model

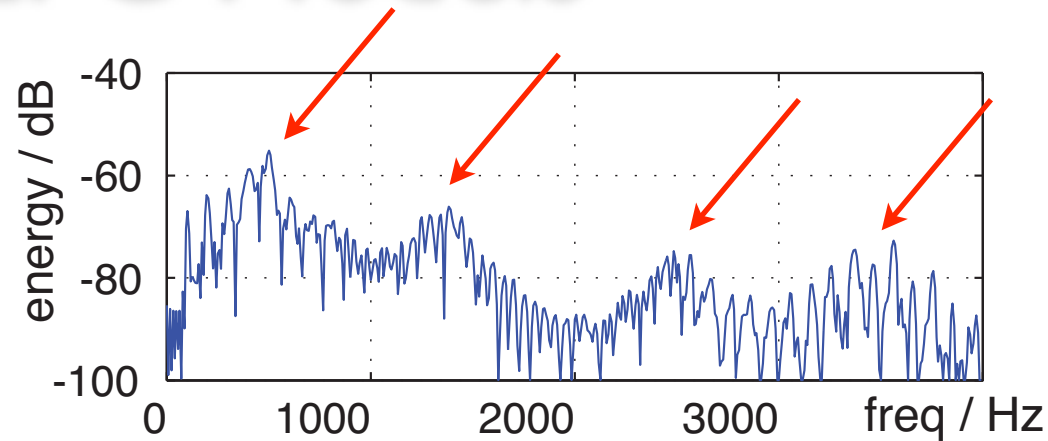
$$s[n] = \sum_{k=1}^p a_k s[n - k] + e[n]$$



- **acoustic tube** model of vocal tract is all-pole
- vocal tract resonances change **slowly** $\sim 10\text{-}20\text{ms}$
- but: **nasals**

Estimating LPC Models

- You can “see” **resonances** in a spectral slice:

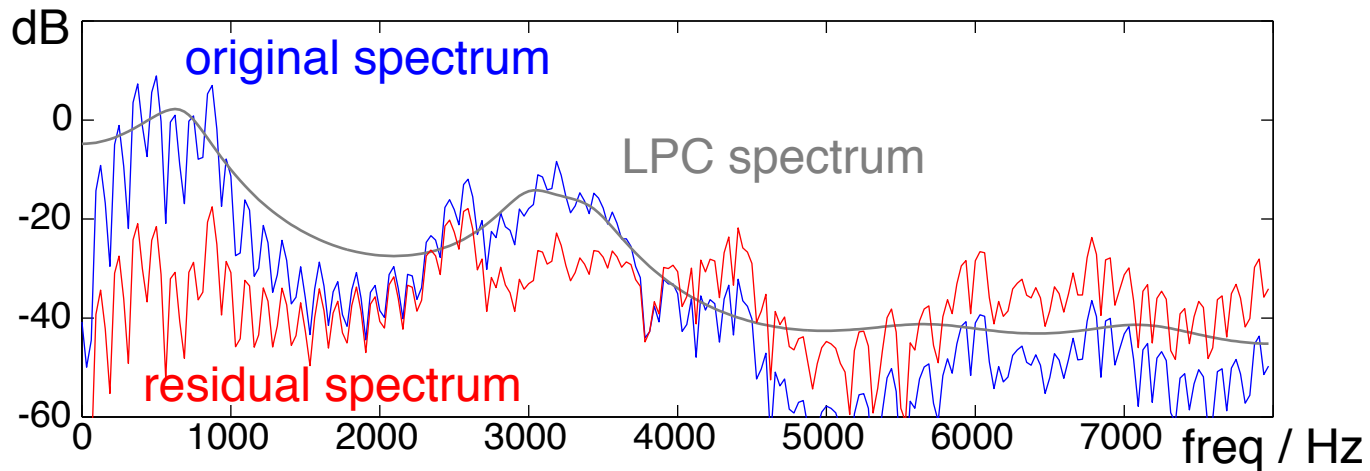
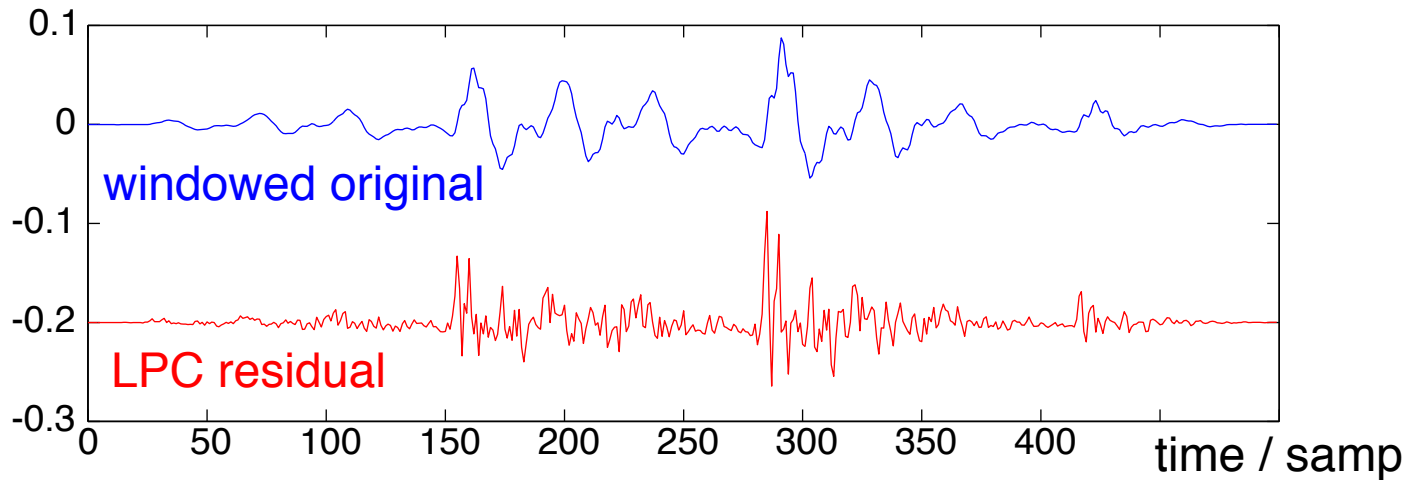


- We can find LPC coefficients $\{a_k\}$ to **minimize energy** of residual $e[n]$:

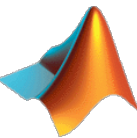
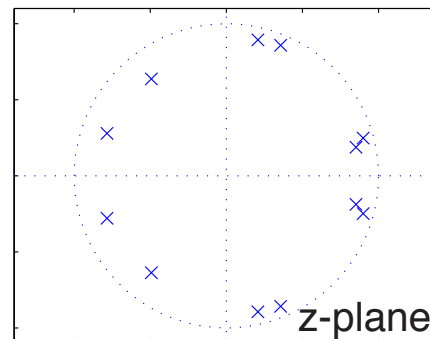
$$\sum_n e^2[n] = \sum_n \left(s[n] - \sum_{k=1}^p a_k s[n-k] \right)^2$$

- differentiate w.r.t. a_k & solve
- end up with p **linear equations** involving **autocorrelations** $r_{ss}(|j-k|) = \sum_n s[n-j]s[n-k]$

LPC Illustration

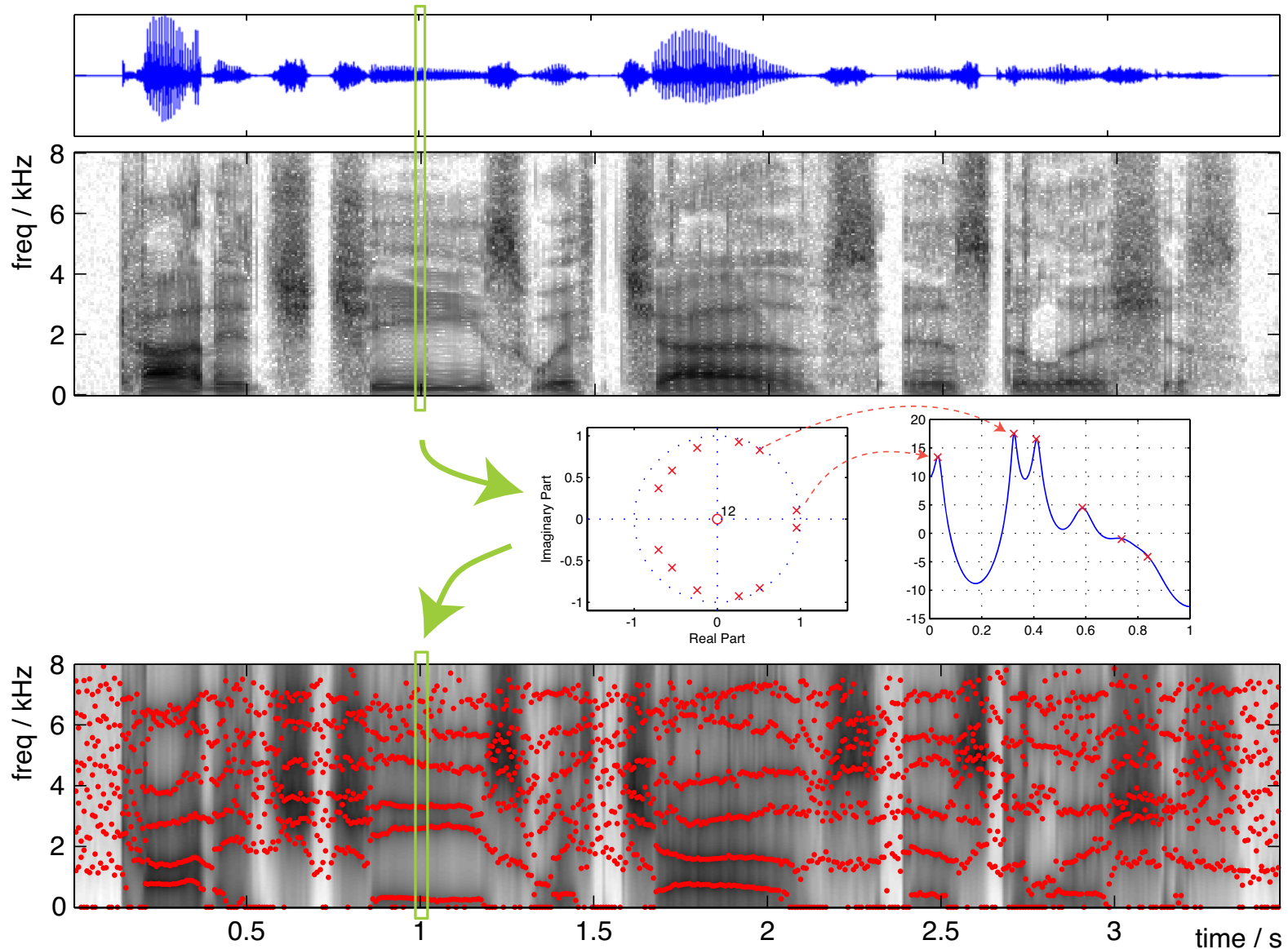


- Actual poles:



Short-Time LP Analysis

- Solve LPC for each ~ 20 ms frame



3. LP Representations

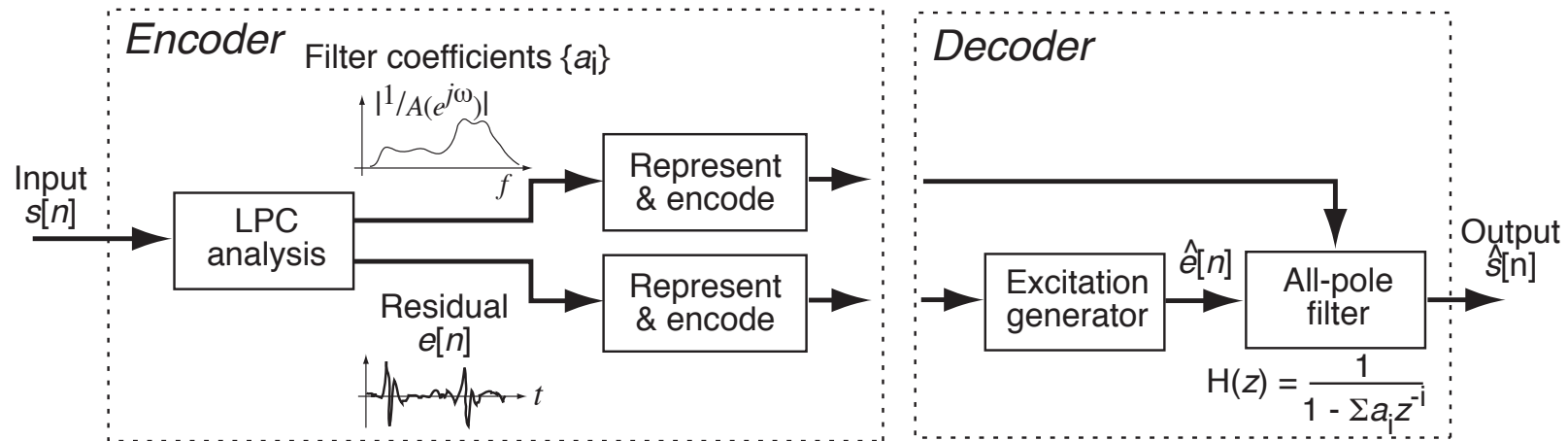
- Can **interpret** LPC filter fit many ways:
- Picking out **resonances**
 - if signal was **source** + **resonances**, should find them
- **Low-order spectral approximation**
 - minimizing $e^2[n]$ also minimizes $|E(e^{j\omega})|^2$
 - different from e.g. Fourier approximation...
- **Finding & removing smooth spectrum**
 - $\frac{1}{A(z)}$ is smooth approximation of $S(z)$
 - $\frac{S(z)}{E(z)} = \frac{1}{A(z)} \Rightarrow E(z) = S(z)A(z)$ is “**unsmoothed**” $S(z)$
- **Signal whitening**
 - removing linear dependence makes residual like **white noise** (iid, flat spectrum)

Alternative Forms

- Many formulations for p^{th} order all-pole IIR:
 - predictor coefficients $\{a_k\}$ / polynomial $A(z)$
 - roots $\{\lambda_i = r_i e^{j\omega_i}\}$ of $A(z)$
 - reflection coefficients (for lattice filter structure)
 - Line Spectral Frequencies (LSF)
- Choice depends on:
 - mathematical convenience
 - numerical stability
 - statistical properties (e.g. for coding)
 - opportunities for modification

4. LPC Synthesis

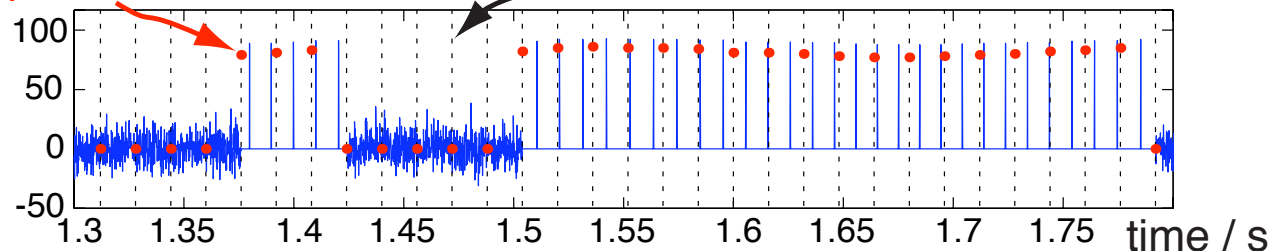
- LP analysis on $\sim 20\text{ms}$ frames gives
 - prediction filter $A(z)$ and residual $e[n]$
 - recombining them should yield perfect $s[n]$
 - coding applications further compress $e[n]$



- e.g. simple pitch tracker \rightarrow “buzz-hiss” encoding

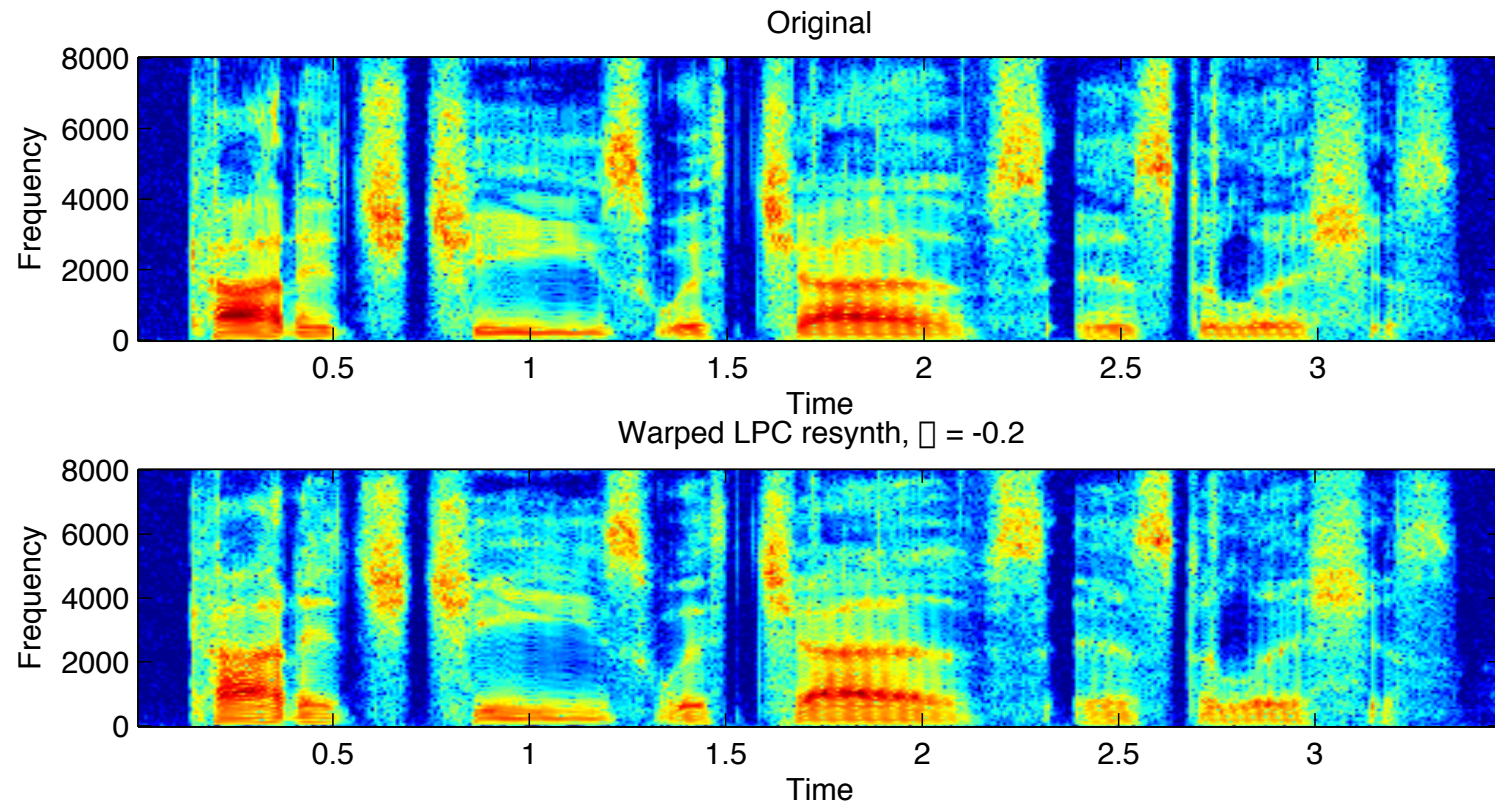
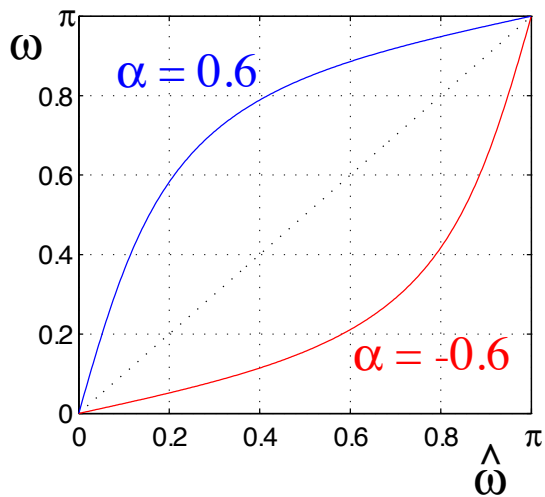
Pitch period values

16 ms frame boundaries



LPC Warping

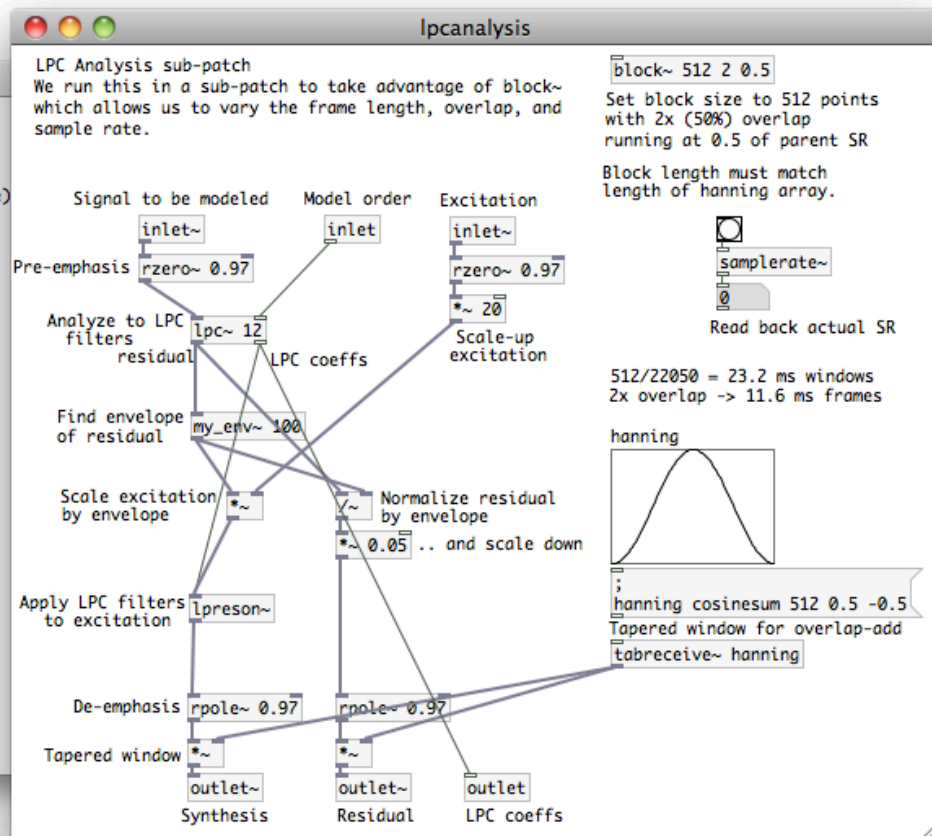
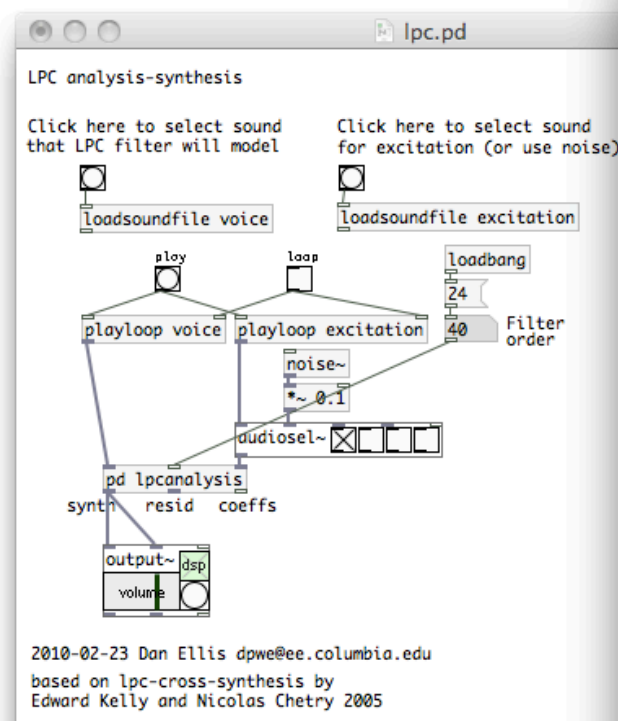
- Replacing delays z^{-1} with **allpass** elements $\frac{z + \alpha}{\alpha z + 1}$ warps frequencies but not magnitudes



◦ <http://www.ee.columbia.edu/~dpwe/resources/matlab/polewarp/>

Cross-Synthesis

- Mix **residual (source)** of one signal with **resonances (filter)** of another
 - or: just use **white noise** as excitation
 - formants carry phonemes → **vocoder**



Summary

- **Resonances** (poles) color sound
- **Source** + **Filter** model
decouples excitation and resonances
- **Linear Prediction** is a simple way to model
and implement resonances (**filter**)
- Many interpretations, representations,
modifications