

# Lecture 2: Linear Prediction

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ECE 417: Multimedia Signal Processing, Fall 2021

- 1 Review: All-Pole Filters
- 2 Inverse Filtering
- 3 Linear Prediction
- 4 Finding the Linear Predictive Coefficients
- 5 Summary

# Outline

- 1 Review: All-Pole Filters
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# All-Pole Filter

An all-pole filter has the system function:

$$H(z) = \frac{1}{(1 - p_1 z^{-1})(1 - p_1^* z^{-1})} = \frac{1}{1 - a_1 z^{-1} - a_2 z^{-2}},$$

so it can be implemented as

$$y[n] = x[n] + a_1 y[n - 1] + a_2 y[n - 2]$$

where

$$a_1 = (p_1 + p_1^*) = 2e^{-\sigma_1} \cos(\omega_1)$$

$$a_2 = -|p_1|^2 = -e^{-2\sigma_1}$$

# Frequency Response of an All-Pole Filter

We get the magnitude response by just plugging in  $z = e^{j\omega}$ , and taking absolute value:

$$|H(\omega)| = |H(z)|_{z=e^{j\omega}} = \frac{1}{|e^{j\omega} - p_1| \times |e^{j\omega} - p_1^*|}$$

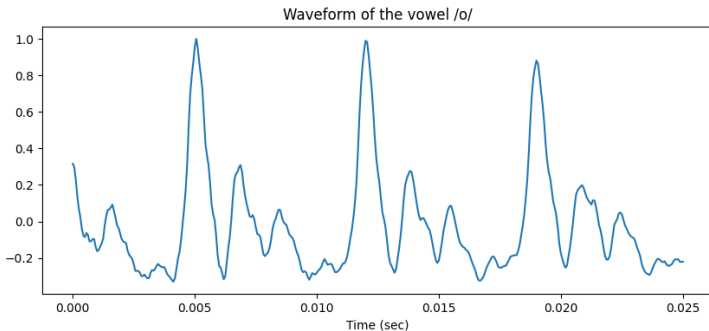
# Impulse Response of an All-Pole Filter

We get the impulse response using partial fraction expansion:

$$\begin{aligned} h[n] &= (C_1 p_1^n + C_1^* (p_1^*)^n) u[n] \\ &= \frac{1}{\sin(\omega_1)} e^{-\sigma_1 n} \sin(\omega_1(n+1)) u[n] \end{aligned}$$

# Speech is made up of Damped Sinusoids

Resonant systems, like speech, trumpets, and bells, are made up from the series combination of second-order all-pole filters.



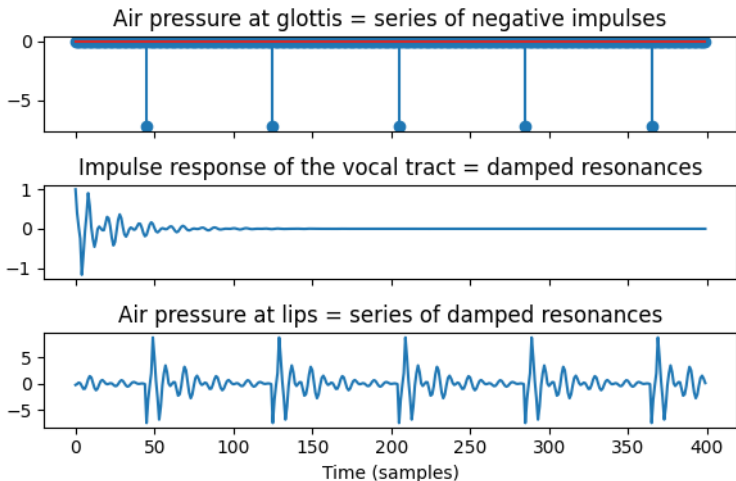
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# Speech

Speech is made when we take a series of impulses, one every 5-10ms, and filter them through a resonant cavity (like a bell).



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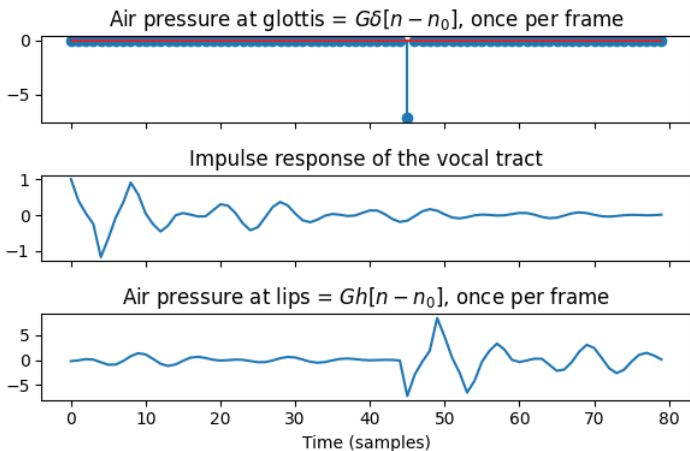
$$S(z) = H(z)E(z) = \frac{1}{A(z)}E(z)$$

where the excitation signal is a set of impulses, maybe only one per frame:

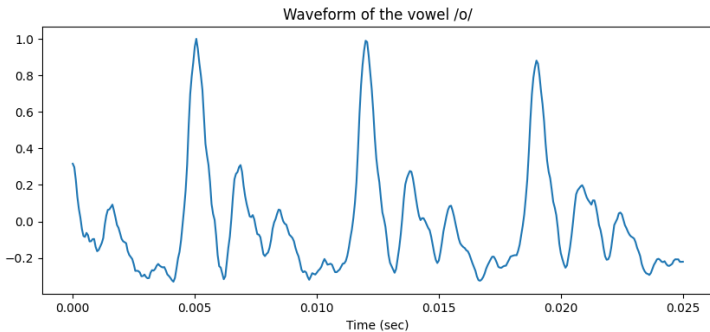
$$e[n] = G\delta[n - n_0]$$

The only thing we don't know, really, is the amplitude of the impulse ( $G$ ), and the time at which it occurs ( $n_0$ ). Can we find out?

# Speech: The Model



# Speech: The Real Thing



# Inverse Filtering

If  $S(z) = E(z)/A(z)$ , then we can get  $E(z)$  back again by doing something called an **inverse filter**:

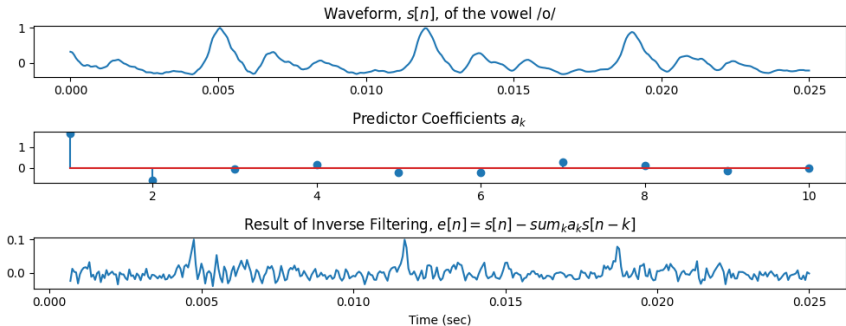
$$\text{IF: } S(z) = \frac{1}{A(z)} E(z) \quad \text{THEN: } E(z) = A(z)S(z)$$

The inverse filter,  $A(z)$ , has a form like this:

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

where  $p$  is twice the number of resonant frequencies. So if speech has 4-5 resonances, then  $p \approx 10$ .

# Inverse Filtering



# Inverse Filtering

This one is an all-pole (feedback-only) filter:

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

That means this one is an all-zero (feedforward only) filter:

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

which we can implement just like this:

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

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# Linear Predictive Analysis

This particular feedforward filter is called **linear predictive analysis**:

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

It's kind of like we're trying to predict  $s[n]$  using a linear combination of its own past samples:

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

and then  $e[n]$ , the glottal excitation, is the part that can't be predicted:

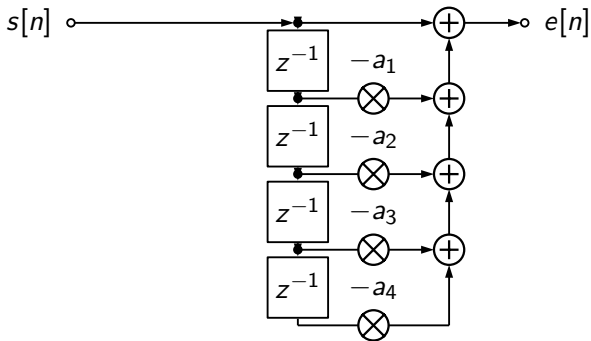
$$e[n] = s[n] - \hat{s}[n]$$

# Linear Predictive Analysis

Actually, linear predictive analysis is used a lot more often in finance, these days, than in speech:

- In finance: detect important market movements = price changes that are not predictable from recent history.
- In health: detect EKG patterns that are not predictable from recent history.
- In geology: detect earthquakes = impulses that are not predictable from recent history.
- ... you get the idea...

# Linear Predictive Analysis Filter



# Linear Predictive Synthesis

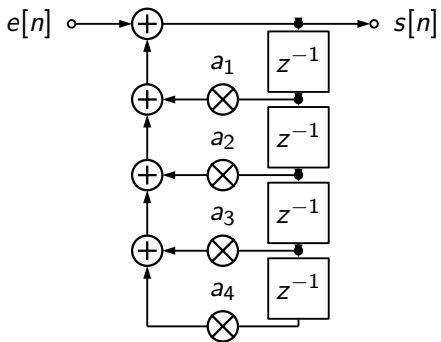
The corresponding feedback filter is called **linear predictive synthesis**. The idea is that, given  $e[n]$ , we can resynthesize  $s[n]$  by adding feedback, because:

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

means that

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

# Linear Predictive Synthesis Filter



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# Finding the Linear Predictive Coefficients

Things we don't know:

- The timing of the unpredictable event ( $n_0$ ), and its amplitude ( $G$ ).
- The coefficients  $a_k$ .

It seems that, in order to find  $n_0$  and  $G$ , we first need to know the predictor coefficients,  $a_k$ . How can we find  $a_k$ ?

# Finding the Linear Predictive Coefficients

Let's make the following assumption:

- Everything that can be predicted is part of  $\hat{s}[n]$ . Only the unpredictable part is  $e[n]$ .



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- So we define  $e[n]$  to be:

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

- ...and then choose  $a_k$  to make  $e[n]$  as small as possible.

$$a_k = \operatorname{argmin} \sum_{n=-\infty}^{\infty} e^2[n]$$

# Finding the Linear Predictive Coefficients

So we've formulated the problem like this: we want to find  $a_k$  in order to minimize:

$$\mathcal{E} = \sum_{n=-\infty}^{\infty} e^2[n] = \sum_{n=-\infty}^{\infty} \left( s[n] - \sum_{m=1}^p a_m s[n-m] \right)^2$$

# Finding the Linear Predictive Coefficients

We want to find the coefficients  $a_k$  that minimize  $\mathcal{E}$ . We can do that by differentiating, and setting the derivative equal to zero:

$$\frac{d\mathcal{E}}{da_k} = 2 \sum_{n=-\infty}^{\infty} \left( s[n] - \sum_{m=1}^p a_m s[n-m] \right) s[n-k], \quad \text{for all } 1 \leq k \leq p$$

$$0 = \sum_{n=-\infty}^{\infty} \left( s[n] - \sum_{m=1}^p a_m s[n-m] \right) s[n-k], \quad \text{for all } 1 \leq k \leq p$$

This is a set of  $p$  different equations (for  $1 \leq k \leq p$ ) in  $p$  different unknowns ( $a_k$ ). So it can be solved.

# Autocorrelation

In order to write the solution more easily, let's define something called the "autocorrelation,"  $R[m]$ :

$$R[m] = \sum_{n=-\infty}^{\infty} s[n]s[n-m]$$

In terms of the autocorrelation, the derivative of the error is

$$0 = R[k] - \sum_{m=1}^p a_m R[k-m] \quad \forall 1 \leq k \leq p$$

or we could write

$$R[k] = \sum_{m=1}^p a_m R[k-m] \quad \forall 1 \leq k \leq p$$

# Matrices

Since we have  $p$  linear equations in  $p$  unknowns, let's write this as a matrix equation:

$$\begin{bmatrix} R[1] \\ R[2] \\ \vdots \\ R[p] \end{bmatrix} = \begin{bmatrix} R[0] & R[1] & \cdots & R[p-1] \\ R[1] & R[0] & \cdots & R[p-2] \\ \vdots & \vdots & \ddots & \vdots \\ R[p-1] & R[p-2] & \cdots & R[0] \end{bmatrix} \begin{bmatrix} a_1 \\ a_2 \\ \vdots \\ a_p \end{bmatrix}$$

where I've taken advantage of the fact that  $R[m] = R[-m]$ :

$$R[m] = \sum_{n=-\infty}^{\infty} s[n]s[n-m]$$

# Matrices

Since we have  $p$  linear equations in  $p$  unknowns, let's write this as a matrix equation:

$$\vec{\gamma} = R\vec{a}$$

where

$$\vec{\gamma} = \begin{bmatrix} R[1] \\ R[2] \\ \vdots \\ R[p] \end{bmatrix}, \quad R = \begin{bmatrix} R[0] & R[1] & \cdots & R[p-1] \\ R[1] & R[0] & \cdots & R[p-2] \\ \vdots & \vdots & \ddots & \vdots \\ R[p-1] & R[p-2] & \cdots & R[0] \end{bmatrix}.$$

# Matrices

Since we have  $p$  linear equations in  $p$  unknowns, let's write this as a matrix equation:

$$\vec{\gamma} = R\vec{a}$$

and therefore the solution is

$$\vec{a} = R^{-1}\vec{\gamma}$$

# Finding the Linear Predictive Coefficients

So here's the way we perform linear predictive analysis:

- 1 Create the matrix  $R$  and vector  $\vec{\gamma}$ :

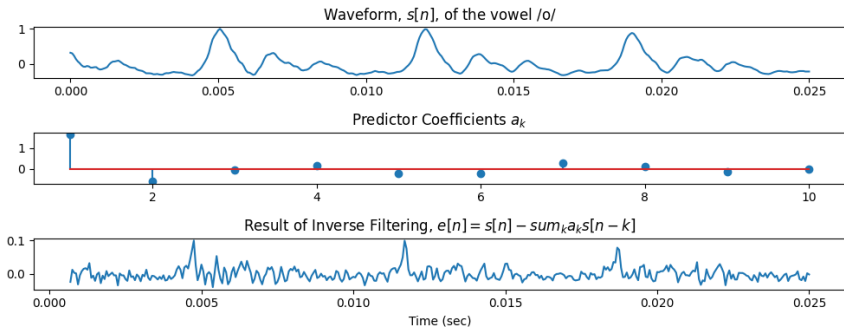
$$\vec{\gamma} = \begin{bmatrix} R[1] \\ R[2] \\ \vdots \\ R[p] \end{bmatrix}, \quad R = \begin{bmatrix} R[0] & R[1] & \cdots & R[p-1] \\ R[1] & R[0] & \cdots & R[p-2] \\ \vdots & \vdots & \ddots & \vdots \\ R[p-1] & R[p-2] & \cdots & R[0] \end{bmatrix}$$

- 2 Invert  $R$ .

$$\vec{a} = R^{-1}\vec{\gamma}$$



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$$\text{IF: } S(z) = \frac{1}{A(z)}E(z) \quad \text{THEN: } E(z) = A(z)S(z)$$

which we implement using a feedforward difference equation, that computes a linear prediction of  $s[n]$ , then finds the difference between  $s[n]$  and its linear prediction:

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

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- So we define  $e[n]$  to be:

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

- ... and then choose  $a_k$  to make  $e[n]$  as small as possible.

$$a_k = \operatorname{argmin} \sum_{n=-\infty}^{\infty} e^2[n]$$

which, when solved, gives us the simple equation  $\vec{a} = R^{-1}\vec{\gamma}$ .