

Linear Predictive Coding

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Idea The speech sample values can be approximated by the linear combination of the past sample values

Model

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

- $s[n]$: Speech output
- a_k : Linear predictive coefficient
- $u_g[n]$: Vocal tract input

Also named "Autoregression(AR) Model"

Theoretical Derivation

Define p-order linear predictor:

$$\tilde{s}[n] = \sum_{k=1}^p \alpha_k s[n-k]$$

corresponding z transformation:

$$\tilde{S}(z) = P(z)S(z)$$

and predictive filter:

$$P(z) = \sum_{k=1}^p \alpha_k z^{-k}$$

Predictive error:

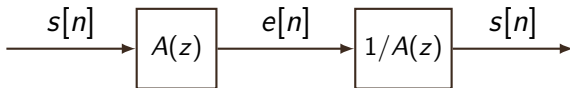
$$\begin{aligned}e[n] &= s[n] - \tilde{s}[n] \\&= \sum_{k=1}^p a_k s[n-k] + Au_g[n] - \sum_{k=1}^p \alpha_k s[n-k] \\&\approx Au_g[n] \text{ (when } \alpha_k \approx a_k \text{)}\end{aligned}$$

corresponding z transformation:

$$E(z) = S(z)(1 - P(z)) = S(z)A(z)$$

and predictive error filter:

$$A(z) = 1 - P(z)$$



MATLAB Code Example

```
1  [x,fs] = audioread('C6_2_y.wav');
2  LENGTH = length(x);
3  n = 0:1/fs:(LENGTH - 1)/fs;
4
5  subplot(3,1,1),plot(n*1000, x),grid
6  xlabel('Time/s'); ylabel('Amplitude');
7  title('Original Signal')
8
9  subplot(3,1,2)
10 order = 20;
11 [a,g] = lpc(x,order);
12 error = filter(a,1,x);
13 plot(n*1000,error, 'r')
14 xlabel('Time/s'); ylabel('Amplitude');
15 title('Predictive Error')
16
17 est_x = filter(1,a,error);
18 subplot(3,1,3), plot(n*1000,est_x), grid;
19 xlabel('Time/s'); ylabel('Amplitude');
20 title('Predictive Signal')
```

MATLAB Plot Example

