

# ECE 537 Fundamentals of Speech Processing

## Problem Set 5

UNIVERSITY OF ILLINOIS  
Department of Electrical and Computer Engineering

Assigned: Monday, 10/3/2022; Due: Monday, 10/10/2022  
Reading: Atal, "Predictive Coding of Speech at Low Bit Rates," 1982

1. Equation (4) in the article gives a formula for the prediction sum-squared error (a.k.a. the energy of the prediction residual) of an  $(m - 1)$ -tap linear prediction. To see why this is the case, let's explore the origin of some of the equations in this section.

(a) (1 point) Define  $\epsilon_m$  in the following way:

$$\epsilon_m = \sum_{n=p+1}^{N+p} (d_n^{(m-1)})^2,$$

where

$$d_n^{(m)} = s_n - \sum_{k=1}^{m-1} a_k^{(m-1)} s_{n-k},$$

and  $a_k^{(m)}$  is the  $k^{\text{th}}$  coefficient of an order- $m$  linear predictor. Solve for  $\frac{\partial \epsilon_m}{\partial a_k^{(m-1)}}$  in terms of the coefficients  $a_i^{(m)}$  and the speech signal  $s_n$ .

- (b) (1 point) Consider setting  $\frac{\partial \epsilon_m}{\partial a_k^{(m-1)}} = 0$  simultaneously for all  $k \in \{1, \dots, m\}$ ; this results in  $m$  linear equations in terms of  $s_n$  and  $a_i^{(m-1)}$ . Convert these  $m$  linear equations into a single matrix equation in terms of the vector  $\vec{a}_{m-1} = [a_1^{(m-1)}, \dots, a_m^{(m-1)}]^T$ , the vector  $\vec{c}_{m-1} = [\phi(0, 1), \dots, \phi(0, m - 1)]^T$ , and the matrix  $\Phi_{m-1}$  defined as

$$\Phi_{m-1} = \begin{bmatrix} \phi(1, 1) & \cdots & \phi(1, m-1) \\ \vdots & \ddots & \vdots \\ \phi(m-1, 1) & \cdots & \phi(m-1, m-1) \end{bmatrix},$$

where

$$\phi(i, j) = \sum_{n=p+1}^{N+p} s_{n-i} s_{n-j}$$

- (c) (1 point) Notice that the equation  $\frac{\partial \epsilon_m}{\partial a_k^{(m-1)}} = 0$  can be written as

$$\sum_{n=p+1}^{N+p} d_n^{(m-1)} s_{n-k} = 0 \tag{1}$$

Eq. (1) is called the orthogonality condition. It says that the coefficient  $a_k^{(m-1)}$  that minimizes  $\epsilon_m$  is the one that eliminates all correlation between  $d_n^{(m-1)}$  (the prediction residual) and  $s_{n-k}$

(the predictor). Use the orthogonality condition to write  $\epsilon_m$  as an affine function of the coefficients  $a_i^{(m-1)}$ , where the coefficients in the linear function are the covariance terms  $\phi(i, j)$ . If your equation has any terms that are quadratic in  $a_i^{(m-1)}$ , then you haven't simplified it far enough, keep going.

- (d) (1 point) The covariance LPC method solves the equation  $\Phi \vec{a} = \vec{c}$  in three steps. First, it computes the Cholesky decomposition  $\Phi = LL^T$ , where  $L$  is a lower-triangular matrix. The Cholesky decomposition is an  $\mathcal{O}\{p^3\}$  operation. Second, it solves for the vector  $\vec{q}$  in the equation  $L\vec{q} = \vec{c}$ ; this is an  $\mathcal{O}\{p^2\}$  operation. Third, it solves for  $\vec{a}$  either directly, by solving the equation  $\vec{q} = L^T \vec{a}$ , or indirectly, using the “known relation between the partial correlations and the predictor coefficients” that is named but not described in the article; in either case, this is an  $\mathcal{O}\{p^2\}$  operation.

The part of all this that's interesting to us is that, if we define the order- $m$  equations as  $\Phi_{m-1} = L_{m-1}L_{m-1}^T$ ,  $L_{m-1}\vec{q}_{m-1} = \vec{c}_{m-1}$ , and  $\vec{q}_{m-1} = L_{m-1}^T \vec{a}_{m-1}$ , then we get that

$$\sum_{i=1}^{m-1} q_i^2 = \vec{q}_{m-1}^T \vec{q}_{m-1} = \vec{a}_{m-1}^T \Phi_{m-1} \vec{a}_{m-1} = \vec{a}_{m-1}^T \vec{c}_{m-1} \quad (2)$$

Use Eq. (2) to re-write your solution to part (c) in terms of the coefficients  $q_i$ .

Digression: here is an observation that is not necessary to solve this problem, but that might help you to deepen your understanding of LPC. In the Cholesky decomposition  $\Phi = LL^T$ , the  $m^{\text{th}}$  row of  $L$  only depends on the first  $m$  rows and columns of  $\Phi$ , therefore  $L_{m-1}$  is a submatrix of  $L_m$ . Similarly, in the equation  $L\vec{q} = \vec{c}$ , the  $m^{\text{th}}$  coefficient,  $q_m$ , depends only on the first  $m$  rows of  $L$ , and the first  $m$  elements of  $\vec{c}$ , therefore the vector  $\vec{q}_{m-1}$  is a subvector of  $\vec{q}_m$ . The same is not true of the equation  $L^T \vec{q}_m = \vec{a}_m$ . The  $i^{\text{th}}$  predictor coefficient of an order- $m$  predictor, is therefore **not the same** as the  $i^{\text{th}}$  coefficient of an order- $p$  predictor:

$$a_i^{(m)} \neq a_i^{(p)},$$

but the partial correlation coefficient  $q_i/\epsilon_i$  is the same for any order,  $m$  or  $p$ , as long as  $m \geq i$  and  $p \geq i$ .

2. (1 point) Suppose that  $d_n$  is the LPC residual, and  $v_n$  is the pitch prediction residual, thus

$$v_n = d_n - \beta_1 d_{n-M+1} - \beta_2 d_{n-M} - \beta_3 d_{n-M-1} \quad (3)$$

If  $d_n$  were perfectly periodic with a period of  $M$ , then it would be possible to set  $v_n = 0$  (after the first pitch period) by simply choosing  $\beta_1 = 0, \beta_2 = 1, \beta_3 = 0$ . The reason Eq. (3) contains three delay terms, instead of just one, is that the pitch period might not be an integer.

Suppose that  $d_n$  is perfectly periodic, but with a period  $\tau$  that is not an integer. In this case, the ideal pitch predictor should be  $P_d(e^{j\omega}) = e^{-j\omega\tau}$  in the range  $-\pi < \omega < \pi$ . What is the inverse transform,  $p_d[n]$ , of this pitch predictor? What would be reasonable values to choose for  $M, \beta_1, \beta_2,$  and  $\beta_3$ ?

3. The idea of perceptual noise shaping is to shape the speech signal, producing  $Y(\omega) = (1 - R(\omega))S(\omega)$ , prior to quantizing it. Quantization generates a synthetic output,  $\hat{q}_n$ , in order to minimize

$$\epsilon = \sum_{n=p+1}^{N+p} q_n^2 = \sum_{n=p+1}^{N+p} (y_n - \hat{y}_n)^2,$$

where

$$\hat{Y}(\omega) = \frac{1}{1 - P_A(\omega)} \hat{Q}(\omega)$$

$$\hat{S}(\omega) = \frac{1}{1 - R(\omega)} \hat{Y}(\omega)$$

- (a) (1 point) Use Parseval's theorem to express  $\epsilon$  as an integral, over frequency, of some function of  $S(\omega)$ ,  $\hat{S}(\omega)$ , and  $R(\omega)$ .
- (b) (1 point) Usually, the quantization noise added in one time step,  $q_n = \hat{y}_n - y_n$ , is uncorrelated with the quantization noise added in any other time step, thus  $q_n$  is white noise, with some average power  $\sigma_q^2$ . Under the assumption that  $q_n$  is white noise with power  $\sigma_q^2$ , what is the power spectrum of  $\hat{s}_n - s_n$ ?
- (c) (1 point) For noise shaping, a reasonable set of principles might include:

1. Near a spectral pole of  $S(\omega)$ , it's OK to have louder noise, because the noise will be masked by the high energy of  $S(\omega)$ , thus the perceptual weighting  $|1 - R(\omega)|^2$  can be smaller at these frequencies, perhaps something like

$$1 \geq |1 - R(\omega)|^2 \geq \frac{1}{|S(\omega)|^2} \text{ if } \omega \approx \omega_k,$$

where  $\omega_k$  is one of the spectral peaks of  $S(\omega)$ .

2. The perceptual weighting should be constant at frequencies far from any spectral pole, thus

$$|1 - R(\omega)|^2 \approx 1 \text{ if } |\omega - \omega_k| \text{ is large}$$

Principle #2 is satisfied if  $1 - R(\omega)$  is an all-pass filter, i.e., its zeros have the same frequencies as its poles. Principle #1 is satisfied if the zeros and poles both have the frequencies of the LPC predictor,  $1 - P_A(z) = 1 - \sum_{k=1}^p a_k z^{-k}$ , and if the bandwidths of the poles are larger than the bandwidths of the zeros. To see that this is the case, consider the all-pass filter

$$1 - R(z) = \frac{1 - p_1 z^{-1}}{1 - a p_1 z^{-1}}, \quad (4)$$

where  $p_1 = e^{-\sigma_1 + j\omega_1}$ ,  $\sigma_1 > 0$  is the (real-valued) half-bandwidth of the pole (measured in radians/sample), and  $a$  is some real constant in the range  $0 \leq a < 1$ . Show that  $|1 - R(e^{j\omega_1})| < 1$ , where  $|1 - R(e^{j\omega_1})| < 1$  is the magnitude response of the all-pass filter at the frequency  $\omega = \omega_1$ .

- (d) (1 point) Suppose that a speech signal has pole frequencies that are measured in radians/sample as  $\omega_k$ , for  $1 \leq k \leq p$ , and corresponding bandwidths of  $2\sigma_k$ . (Assume that these are arranged in complex conjugate pairs, e.g.,  $\omega_{p+1-k} = -\omega_k$ , and  $\sigma_{p+1-k} = \sigma_k$ ). The LPC polynomial is therefore

$$1 - P_A(z) = 1 - \sum_{k=1}^p a_k z^{-k} = \prod_{i=1}^p (1 - p_i z^{-1}),$$

where  $p_i = e^{-\sigma_i + j\omega_i}$ . Eq. (21) in the article suggest using a perceptual weighting filter that has  $1 - P_A(z)$  in the numerator, and the following denominator:

$$1 - \sum_{k=1}^p \alpha^k a_k z^{-k}, \quad (5)$$

where  $0 \leq \alpha \leq 1$ . Show that Eq. (5) has roots that have the same frequencies ( $\omega_i$ ) as  $1 - P_A(z)$ , and that its bandwidths have been increased to  $\sigma_i - \ln \alpha$ .