

## EE 627 - Speech Signal Processing

### Assignment # 5

1. Prove that for a LPC model described by  $Q(z) = 1 - a_1z^{-1} - a_2z^{-2} - \dots - a_nz^{-n}$ , The poles migrate towards the unit circle as  $a_n \rightarrow \infty$ . Sketch the same on the unit circle.
2. Show that the phase derivative  $\angle X(\omega)$  of the Fourier transform  $X(\omega)$  of a sequence  $x[n]$  can be obtained through the real and imaginary parts of  $X(\omega)$ ,  $X_r(\omega)$  and  $X_i(\omega)$ , respectively, as

$$\angle X(\omega) = \frac{X_r(\omega) \bar{X}_i(\omega) - X_i(\omega) \bar{X}_r(\omega)}{|X(\omega)|^2}$$

Where  $|X(\omega)|$ , the Fourier transform magnitude of  $x[n]$ , is assumed non-zero.

3. (MATLAB) In this problem, you use the speech waveform *speech1\_10k* (at 10000samples/s) in the workspace *ex6M.mat*. The exercise works through a problem in homomorphic deconvolution, leading to the method of homomorphic prediction.

- a) Window *speech1\_10k* with a 25-ms Hamming window. Using a 1024-point FFT, compute the real cepstrum of the windowed signal and plot. For a clear view of the real cepstrum, set the first cepstral value to zero (which is the DC component of the log-spectral magnitude) and plot only the first 256 cepstral values.
- b) Estimate the pitch period (in samples and in milliseconds) from the real cepstrum by locating a distinct peak in the quefrency region.
- c) Extract the first 50 low-quefrency real cepstral values using a lifter of the form

$$l[n] = \begin{cases} 1, & n = 0 \\ 2, & 1 \leq n \leq 49 \\ 0, & \text{otherwise} \end{cases}$$

Then Fourier transform (using a 1024-point FFT) and plot the first 512 samples of the resulting log-magnitude and phase.

- d) Compute and plot the minimum-phase impulse response using your result from part (c). Plot just the first 200 samples to obtain a clear view. Does the impulse response resemble one period of the original waveform? If not, then why not?
- e) Use your estimate of the pitch period (in samples) from part (b) to form a periodic unit sample train, thus simulating an ideal glottal pulse train. Make the length of the pulse train 4 pitch periods. Convolve this pulse train with your (200-sample) impulse response estimate from part (d) and plot. You have now synthesized a minimum-phase counterpart to the (possibly mixed-phase) vowel *speech1\_10k*. What are the differences between your construction and the original waveform?