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MANIPAL INSTITUTE OF TECHNOLOGY  
Manipal University



**SIXTH SEMESTER B.TECH (E & C) DEGREE END SEMESTER EXAMINATION**  
**MAY/JUN 2016**  
**SUBJECT: DIGITAL SPEECH PROCESSING (ECE - 326)**

**TIME: 3 HOURS**

**MAX. MARKS: 50**

**Instructions to candidates**

- Answer **ANY FIVE** full questions.
- Missing data may be suitably assumed.

- 1A. Starting from the wave equations derive the expression for the transfer function of uniform lossless tube model.
- 1B. A segment of a speech signal is obtained using Hamming window, whose spectrum is given in **Figure 1B**. Find the corresponding pitch and first formant frequencies.
- 1C. Briefly explain the role of vocal cords in speech production.
- (5+3+2)
- 2A. i) With a block diagram explain the pitch period estimation using simple inverse filtering and tracking method.
- ii) Explain how formant frequencies can be estimated using LPC parameters.
- 2B. Assume that STAMF has to be calculated on a speech signal sampled at 12.8KHz using a Hamming window with length of 31.25ms at a rate of 160 times/second. Find the amount of overlapping between two successive frames.
- 2C. Explain how pitch period can be estimated using short time average magnitude difference function.
- (5+3+2)
- 3A. Consider the following signal  $x[n] = \cos(w_0 n) + \cos(w_1 n) + \cos(w_2 n)$  truncated by a rectangular window  $w[n]$  of length L. Verify the ability to distinguish the frequency components in the signal for the following two cases and comment on the results. Case 1) L=25 2) L=100. Given  $w_0 = 0.2\pi \text{ rad}$ ,  $w_1 = 0.22\pi \text{ rad}$ ,  $w_2 = 0.6\pi \text{ rad}$
- 3B. With a neat block diagram, explain the filter bank procedure for analysis of discrete STFT.
- 3C. Briefly explain how the length of analysis window affects the resolution of the spectrogram.
- (5+3+2)

- 4A. Explain a method of speech vs. silence discrimination using Energy and Zero-crossings.
- 4B. Briefly explain the autocorrelation method of LPC analysis.
- 4C. Consider a first order linear predictor whose output is given by  $y(n) = \alpha s(n - 1)$ .  
Let  $e(n) = s(n) - y(n)$  be the error signal. Derive the expression for the optimum value of  $\alpha$  by minimizing the mean squared error. (5+3+2)
- 5A. What is a Mel filter bank? Explain how MFCC feature vectors are obtained from speech signal.
- 5B. Explain briefly the different methods available for evaluating speech quality.
- 5C. Explain working of feedback adaptive quantiser with time varying gain. (5+3+2)
- 6A. Explain how speech signal can be synthesized using LPC synthesizer. Also explain the working of LPC vocoder.
- 6B. What are the different elements of an HMM?
- 6C. Consider a signal  $x(n) = [-2, -1, -1, 2, 3, 4]$ . If  $R_n(k)$  represents autocorrelation of  $x(n)$ , find  $R_n(0)$ . (5+3+2)

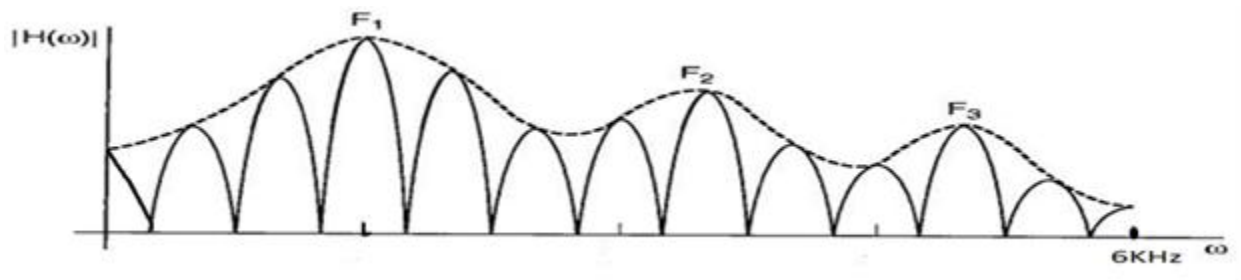


Figure 1B