



MANIPAL INSTITUTE OF TECHNOLOGY
Manipal University
**SIXTH SEMESTER B.TECH (ECE) DEGREE END SEMESTER
EXAMINATION - April/May 2017**
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. Briefly explain the role of vocal tract in speech production.
- 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 a sequence $x(n)$ with DTFT $X(e^{j\omega})$. Let $\tilde{X}(k)$ be the DFT of $x(n)$ obtained by sampling $X(e^{j\omega})$ at frequencies $\omega_k = \frac{2\pi k}{N}$, $k = 0, 1, 2, \dots, N-1$. If $\tilde{x}(n)$ represent the inverse DFT of $\tilde{X}(k)$, show that $\tilde{x}(n) = \sum_{r=-\infty}^{\infty} x(n - rN)$.
- 3B. Consider a speech signal sampled at 12KHz. Find the minimum total sampling rate required to obtain a STFT representation in the following cases.
- i) If a Rectangular window of length 25ms is used.
- ii) If a Hamming window of length 25ms is used.
- 3C. Let $y[n] = x[n] - x[n-1]$. Show that STFT of the given signal can be represented as
- $$Y_n(e^{j\omega}) = X_n(e^{j\omega}) - e^{j\omega} X_{n-1}(e^{j\omega})$$
- (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 α 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 $Rn(k)$ represents autocorrelation of $x(n)$, find $Rn(0)$.
- (5+3+2)