

## Seventh Semester B.E. Degree Examination, Dec.2013/Jan.2014

## Speech Processing

Time: 3 hrs.

Max. Marks:100

Note: Answer FIVE full questions, selecting at least TWO questions from each part.

PART – A

- 1 a. With a schematic, describe the speech production mechanism and identify the source system components. Also explain the classification of speech sound according to mode of excitation. (12 Marks)
- b. What is a phoneme? Explain in detail, the vowels, semi vowels, Diphthongs and Nasals. (08 Marks)
- 2 a. With a related block diagram, equations explain how short time energy and average magnitude of speech signal is computed. (12 Marks)
- b. With the related equations, block diagram explain short time average zero crossing rate in detail. Also list the limitation of short-time-zero crossing detector. (08 Marks)
- 3 a. Explain the operation of pitch period estimation using parallel processing approach. With a related block diagram and graph. (10 Marks)
- b. From the basic equation for autocorrelation function of a discrete-time deterministic signal  $\phi(K)$ , derive the equation for short time auto correlation function  $R_n(K)$ . Draw the related block diagram so as to obtain  $R_n(K)$  from the sequence  $X(n)$ . Also discuss modified auto correlation function with their window condition. (10 Marks)
- 4 a. With respect to temporal processing of speech signal. Explain the following :
  - (i) Temporal filtering. (10 Marks)
  - (ii) Non-linear transformation of time-trajectories. (10 Marks)
- b. What is RASTA processing? Explain with related block diagram and equations. Also draw the frequency response of the RASTA band pass filter. (10 Marks)

PART – B

- 5 a. Define short-time Fourier transform. Also explain the Fourier transform interpretation of short-time-Fourier-transform (STFT). (10 Marks)
- b. Explain the linear filter operation of a short-time spectrum analysis with related equations and block diagram. Also discuss about magnitude of the short time spectrum using both low pass filters and band pass filter. (10 Marks)
- 6 a. Explain filter bank summation method of short time synthesis in signal in terms of linear filtering. (10 Marks)
- b. Explain overlap addition method for short time synthesis. (10 Marks)
- 7 a. Explain the homomorphic system for convolution with related equation and block diagram. (10 Marks)
- b. With a related block diagram, explain homomorphic vocoder containing an analyzer and synthesizer. (10 Marks)
- 8 a. Write a short note on voice response system. Also explain with a block diagram of an all digital voice response system. (10 Marks)
- b. Write a short note on on-line digital speaker verification system. Also explain with a block diagram of the signal processing aspects of the speaker verification system. (10 Marks)

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