**Code No: C4502** 

## JAWAHARLAL NEHRU TECHNOLOGICAL UNIVERSITY HYDERABAD M.Tech I Semester Examinations, March/April 2011 ADVANCED DIGITAL SIGNAL PROCESSING (SYSTEM & SIGNAL PROCESSING)

Time: 3hours Max. Marks: 60

## Answer any five questions All questions carry equal marks

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- 1. a) Explain clearly the need for sample rate conversion and discuss about different techniques of sample rate conversion.
- b) Explain clearly the procedure for decimation by a factor D and bring and bring out the considerations for selection of LPF. Impulse response  $\ell(n)$ . Show all the steps involved.
- 2. a) What do you understand by the term Polyphase structures for sample rate conversion and discuss the steps for efficient implementation of sampling rate conversion by a factor (I/D).
  - b) Discuss the implementation of multistage sections for decimation of 4KHZ audio signal sampled at 8 KHZ. Given frequency components below 80HZ are to be isolated. Using a filter with pass bound  $0 \le F \le 75HZ$  and transmission band  $75 \le F \le 80HZ$ . The attenuation is pass bound  $8_1 \le 10^{-2}$  and in stop band is  $8_2 \le 10^{-4}$ . Design a multistage F/R filter. [12]
- 3. a) Explain clearly the power spectral estimation of finite discussion signals using period a gram techniques bring out the limiting of it.
  - b) Discuss the implementation of Bartlett Welch methods for improvement of power spectral estimation and compare them. [12]
- 4. a) Discuss the principle of parametric modeling and hence bring out different models used for power spectral estimation and obtain the relation between model parameters and Autocorrelation of data series.
- b) Obtain the simplified relation between the parameters and Auto core ration parameters of AR model of order N and explain the power spectral estimation using this model. [12]
- 5. a) Define Brag Algorithm and Explain the implementation of it and bring out the advantages and disadvantages of it.
  - b) Discuss importance of Model order collection and explain different methods available. [12]
- 6. a) Explain the advantages of Lattice structures for FIR implementation and hence obtain the relation between . Difference equation coefficients representing N th order FIR system and Lattice parameter.
- b) Determine the Lattice structure parameters { Km } per. Implementation if FIR filter with system function.  $H(z) = 1 + 2z^{-1} + z^{-2}$ . [12]

7.a) Explain clearly the effect of Finite word representation in DSP and I	nence
distinguish clearly between truncation and Round of errors.	

b) Derive expression for Round off and truncation errors in case of Fixed point represent ion and Floating point representation.

[12]

## 8. Write short notes on

- a) Sub band coding and QMF filter in multirate signal processing.
- b) Lattice implementation of IIR filters.

[12]

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