

**SECOND SEMESTER M.TECH (DEAC) DEGREE END SEMESTER EXAMINATION
MAY/JUNE 2016**

SUBJECT: ADVANCED DIGITAL SIGNAL PROCESSING (ECE - 504)

TIME: 3 HOURS

MAX. MARKS: 50

Instructions to candidates

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

- 1A. Explain the basic interpolation operation. Derive expression for the spectrum of interpolated signal in terms of original spectrum. What is the role of interpolation filters?
- 1B. For the system shown in Fig.Q1.B, determine $y(n)$ in terms of $x(n)$.

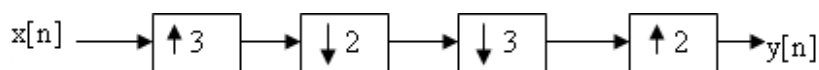


Fig. Q1.B

- 1C. Certain analog signal band limited to 3.3 KHz is sampled at twice the Nyquist rate to get discrete signal $x(n)$. What is the maximum integer by which this can be decimated so that there is no aliasing?

(5+3+2)

- 2A. Explain the principle of IFIR design. It is required to implement 96 KHz to 1 KHz down converter. If the overall decimation filter has $f_p=450$ Hz, $\delta_p=0.01$ dB and $\delta_s=60$ dB, get the specifications of individual filters for IFIR design (Assume interpolation factor of 48, ie. take $96=48 \times 2$).
- 2B. Show that there is no phase distortion in a normal alias-free QMF bank using linear-phase FIR filters. Explain why the filter length should be even.
- 2C. For the system shown in Fig. 2.C, determine the output sequence $y(n)$. All the input signals have same sampling rate.

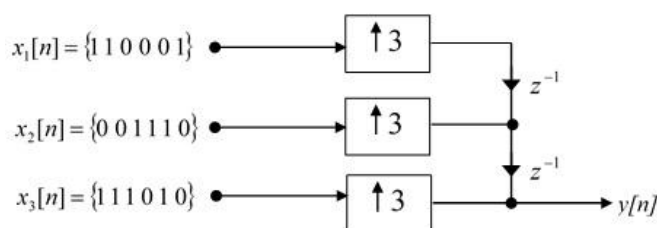


Fig. 2.C

(5+3+2)

- 3A. Describe the uniform DFT filter bank for the computation of N-point DFT. Draw the implementation block diagram of DFT filter bank using poly-phase filters and estimate the cost of such filter bank if $N=32$. The base filter has length 51.
- 3B. Explain the concept of optimum and adaptive filtering. Write the Widrow-Hopf equation for optimum filter and LMS equation for adaptive tapped delay line filters.

- 3C. Draw the convergence characteristics of LMS algorithm. How to choose convergence parameter?
(5+3+2)
- 4A. Describe the concept of optimum predictor. Derive expression for the optimum predictor coefficients for one step, M tap optimum predictor.
- 4B. Describe the functioning of Correlation canceler loop (CCL).
- 4C. Write the implementation block diagram of LMS algorithm.
(5+3+2)
- 5A. Derive expressions for the cepstrum of signals that can be expressed as sum of exponentials.
- 5B. Explain the concept of homo-morphic signal processing. Discuss the signal model for such applications.
- 5C. Draw the block diagram of homo-morphic system for the computation of cepstrum using DFT. How to select the DFT length?
(5+3+2)
- 6A. Certain signal with Z-transfer function $X(z) = \frac{0.95 + z^{-1}}{(1 - 0.8z^{-1} + 0.64z^{-2})}$ is transmitted over multipath channel having impulse response $h(n) = \delta(n) + 0.8\delta(n-10) + 0.64\delta(n-20)$. Determine the cepstrum of the received signal. Explain how the echoes can be suppressed from the received signal.
- 6B. Compare parametric and non- parametric method of PSD estimation.
- 6C. What is the role of autocorrelation function in estimating PSD?
(5+3+2)