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MANIPAL INSTITUTE OF TECHNOLOGY

MANIPAL

A Constituent Institution of Manipal University

V SEMESTER B.Tech.(BME) DEGREE END SEM EXAMINATIONS NOV/DEC 2016

SUBJECT: BIOMEDICAL DIGITAL SIGNAL PROCESSING (BME 309)

(REVISED CREDIT SYSTEM)

Saturday, 3rd December 2016, 2 PM to 5 PM

TIME: 3 HOURS

MAX. MARKS: 100

Instructions to Candidates:

1. Answer any FIVE full questions.
2. Draw labeled diagram wherever necessary

1. (A) Consider two LSI systems with impulse responses $h_1[n] = \delta[n - 2]$ and $h_2[n] = \delta[n - 3]$ respectively. Calculate the overall impulse response $h[n]$, if the systems are connected in (i) cascade (ii) parallel. 08
- (B) Derive the condition (based on the impulse response), for stability and causality of LSI systems. 06
- (C) Use the Z-transform to perform the convolution of the following two sequences: 06

$$h(n) = \begin{cases} \left(\frac{1}{2}\right)^n; & 0 \leq n \leq 2 \\ 0 & \text{elsewhere} \end{cases} \quad x(n) = \delta(n) + \delta(n - 1) + 4\delta(n - 2)$$
2. (A) (i) Find the DFT $X[k]$ of the sequence, $x[n] = \{1, 2, 3, 1\}$ using the Matrix method. 06

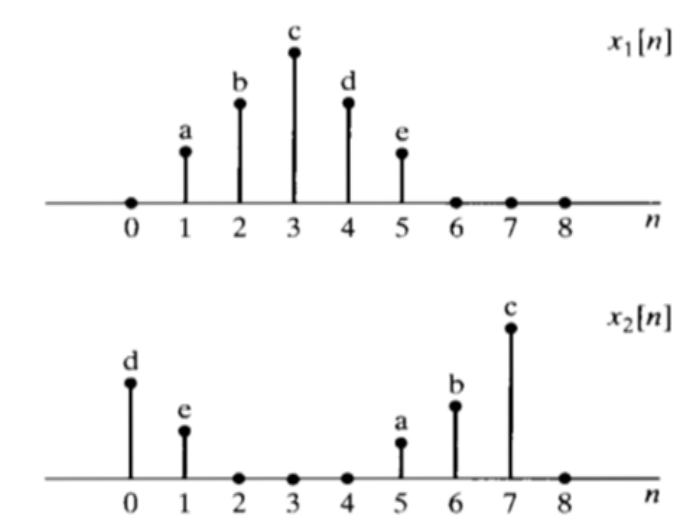
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 (ii) Find the IDFT $x[n]$ from $X[k]$ obtained in part (i) using the Matrix method.
- (B) Design a FIR high pass filter having cutoff frequency 1.2 radians using a hamming window of length $N=7$. 08
- (C) State the condition (based on the ROC of the system function, $H(z)$), for stability and causality of LSI systems. Check for causality and stability of the LSI system defined by the unit sample response $h[n] = a^n u[n]$, using the above condition. 06
3. (A) List the characteristics of FIR filters. 04
- (B) Consider a filter with impulse response $h(n) = \{1, 1, 1\}$. If the input to the filter is $x(n) = \{3, -1, 0, 1, 3, 2, 0, 1, 2, 1\}$, find the output using the overlap-save method of convolution. 08

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- (C) The two 8-point sequences $x_1[n]$ & $x_2[n]$ shown in figure below have DFTs $X_1(k)$ & $X_2(k)$, respectively. Observe the two sequences carefully, and determine the relationship between $X_1(k)$ & $X_2(k)$.

08



4. (A) A continuous-time signal $x(t)$ is obtained at the output of an ideal lowpass filter with cutoff frequency $\omega_c = 1000\pi \text{ rad/sec}$. If impulse-train sampling is performed on $x(t)$, which of the following sampling periods would guarantee perfect recovery of $x(t)$ from its sampled version using an appropriate lowpass filter? 04
- (i) $T = 0.5 \times 10^{-3} \text{ Sec}$ (ii) $T = 2 \times 10^{-3} \text{ Sec}$ (iii) $T = 10^{-4} \text{ Sec}$
- (B) A causal LSI system is characterized by the difference equation, 08
- $$y(n) - \frac{5}{6}y(n-1) + \frac{1}{6}y(n-2) = x(n).$$
- (a) Determine: (i) the system function, $H(z)$ (ii) the unit sample response, $h(n)$
- (b) Determine whether the system is stable. Justify your answer.
- (C) Calculate the 8-point DFT of the sequence $x[n] = \{1, 2, 3, 4, 4, 3, 2, 1\}$, using DIT-FFT 08
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- radix-2 algorithm.
5. (A) Assume that a complex multiplication takes $1\mu\text{s}$ and that the amount of time to compute a DFT is determined by the amount of time it takes to perform all of the multiplications. 04
- (i) Find the time taken to compute a 1024-point DFT directly?
- (ii) What is the computation time if an FFT is used?
- (iii) Repeat parts (i) and (ii) for a 4096-point DFT.

- (B) An analog Chebyshev low pass filter is to be designed to meet the following specifications: 08

Passband ripple: $\leq 3\text{dB}$

Passband edge: 100 rad/sec

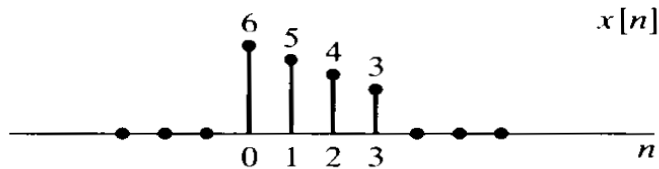
Stopband attenuation: $\geq 25\text{dB}$

Stopband edge: 250 rad/sec

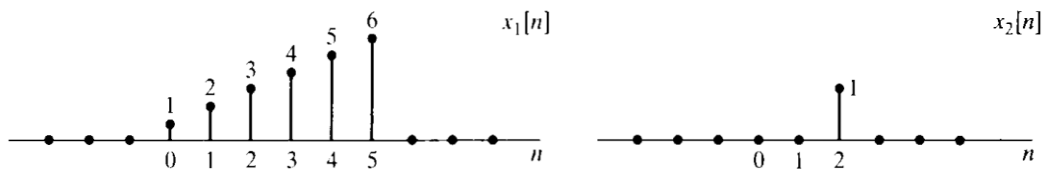
Obtain the Transfer function & Frequency response of the filter.

- (C) (a) Figure shows a finite length sequence $x[n]$. Sketch the following sequences in the range $0 \leq n \leq 3$. 08

(i) $x((n-3))_4$ (ii) $x((n+2))_4$ (iii) $x((-n))_4$



- (b) Figure shows two finite length sequences $x_1[n]$ & $x_2[n]$. Sketch their 6-point circular convolution.



6. (A) What is the necessary and sufficient condition for a discrete-time filter to have linear-phase? An FIR filter is specified by the following impulse response: 06

$$h(n) = -\frac{1}{3}\delta(n) + \frac{1}{2}\delta(n-1) - \frac{1}{3}\delta(n-2).$$

Is this a linear-phase filter?

- (B) A digital Butterworth low pass filter is to be designed using the impulse invariant method to satisfy the following constraints: 08

$$20 \log|H(w)|_{w=0.2\pi} \geq -1.9328 \text{ dB}$$

$$20 \log|H(w)|_{w=0.6\pi} \leq -13.9794 \text{ dB}$$

Obtain the transfer function $H(z)$ of the filter.

- (C) Illustrate the concepts of sampling and quantization with an example each. 06