



SIXTH SEMESTER B.TECH. (INSTRUMENTATION AND CONTROL ENGG.)

END SEMESTER EXAMINATIONS, APRIL - 2018

SUBJECT: DIGITAL SIGNAL PROCESSING [ICE 3202]

Duration: 3 Hour

Max. Marks:50

Instructions to Candidates:

- ❖ Answer **ALL** the questions.
- ❖ Missing data may be suitably assumed.

1A A causal LTI system has following system function

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$$H(z) = \frac{(1 + z^{-1})}{\left(1 - \frac{1}{2}z^{-1}\right)\left(1 + \frac{1}{4}z^{-1}\right)}$$

- i. What is the ROC of $H(z)$? Is this system stable?
- ii. Find z-transform $X(z)$ of an input $x[n]$ that will produce the output $y[n] = (-1/3) (-1/4)^n u[n] - (4/3) (2)^n u[-n-1]$.
- iii. Find the impulse response of the system.

1B Determine the inverse z-transform of $X(z) = \frac{1}{1 - 0.8z^{-1} + 0.12z^{-2}}$

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if ROC is, $0.2 < |z| < 0.6$

1C State and prove the multiplication by n property of z- transform.

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2A By means of DFT and IDFT, determine the response of LTI system with impulse response $h[n] = \{1, 2, 3\}$ to the input $x[n] = \{1, 2, 2, 1\}$

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2B State and prove the circular convolution property of DFT.

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2C Bring out the mathematical relation between DFT and Z transform.

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3A The 8-point DFT of a certain signal is

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$X(k) = \{20, -5.828 - j2.414, 0, -0.171 - j0.414, 0, -0.171 + j0.414, 0, -5.828 + j2.414\}$.
Using DIT-FFT algorithm, find the time domain representation of $X[k]$.

- 3B** Illustrate 4-point radix-2 DIF- FFT algorithm with signal flow diagrams. Discuss on the computational advantage of the algorithm. **3**
- 3C** Compute the DFT of the sequence $x(n) = \{1, j, -1, -j\}$ for $N = 4$. **2**
- 4A** Explain the impulse invariant transformation method of digitizing analog filter. Using this method find $H(z)$ for the filter having transfer function $H(s) = \frac{2}{(s+2)}$. Assume sampling frequency of 1 Hz. Illustrate the position of pole of $H(z)$ in the z plane. **5**
- 4B** Determine the poles of lowpass Butterworth filter for $N=3$. Sketch the location of poles on s -plane and hence determine the normalized transfer function of lowpass filter. **3**
- 4C** Compare the impulse invariant and bilinear transformations (any 4 difference) **2**
- 5A** It is required to have digital FIR filter with following frequency response. **5**

$$H(e^{jw}) = e^{-j3w}; \quad |w| \leq \frac{\pi}{2}$$

$$= 0; \quad \frac{\pi}{2} < |w| < \pi$$

What is the length of this filter? Determine the unit impulse response of this filter using non-recursive frequency sampling technique.

- 5B** What is the need for spectral estimation? Explain any one method used in averaging the periodogram. **3**
- 5C** Compare the FIR and IIR filter (any 4 difference) **2**
