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**FINAL EXAMINATION
SEMESTER I
SESSION 2019/2020**

COURSE NAME : DIGITAL SIGNAL PROCESSING
COURSE CODE : BEB 30503 / BEF 35603
PROGRAMME CODE : BEJ / BEV
EXAMINATION DATE : DECEMBER 2019 / JANUARY 2020
DURATION : 3 HOURS
INSTRUCTION : ANSWER ALL QUESTIONS

THIS QUESTION PAPER CONSISTS OF **FIFTEEN (15)** PAGES

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Q1 (a) The operation of each subsystem in **Figure Q1(a)** are described as below:

- $T_1\{\cdot\}$ produces $y[n] = x\left[\frac{n}{2}\right]$ based on zero interpolation
- $T_2\{\cdot\}$ produces $z[n] = y\left[\frac{n}{3}\right]$ based on step interpolation

Let the input signal, $x[n] = \cos(2n\pi)$;

(i) Sketch the output signal $z[n]$ for $0 \leq n \leq 11$ (4 marks)

(ii) Prove that the output signal can be represented by

$$z[n] = \sum_{p=-\infty}^{\infty} \text{rect}\left(\frac{n - (1 + 6p)}{2}\right)$$

(5 marks)

(b) Calculate the periodic convolution of $y_p[n] = x_p[n] \otimes h_p[n]$ for $x_p[n] = \{-1, 2, 5\}$ and $h_p[n] = \{3, 0, -4\}$ by using cyclic method. (5 marks)

(c) Given the signal $x[n] = \{A, 2, 3, 2, A\}$. Analyze the possible value of A if autocorrelation of $x[n]$ gives $r_{xx}[0] = 19$. Use sum-by-column method for linear convolution process. (6 marks)

- Q2** (a) State **THREE (3)** cases of sampling according to the given condition of sampling rate, S and band-limited frequency, B

- (i) $S = 2B$
- (ii) $S > 2B$
- (iii) $S < 2B$

(3 marks)

- (b) An Analog to Digital Converter (ADC) is consists of 3 modules: (1) sampler, (2) quantizer and (3) coder modules. ADC is used to transform the analog signal into encoded digital signal before it can be transmitted to receiver station using a suitable transmission medium.

The analog signal input of ADC has two different sources which are:

Source A: $x_1(t) = 3 \sin\left(2\pi t + \frac{\pi}{4}\right) - 3 \cos\left(2\pi t + \frac{3\pi}{2}\right) \text{ V}$

Source B: $x_2(t) = 1.5 \cos\left(3\pi t + \frac{3\pi}{4}\right) - \sin(3\pi t) \text{ V}$

Both analog signal is sampled at a sampling rate of 9 Hz with dynamic range of $\pm 2.5 \text{ V}$.

- (i) Find the discrete signal of $x_1[n]$ and $x_2[n]$ for $0 \leq n \leq 3$.

(5 marks)

- (ii) Given the encoded digital signal $x_c[n]$ for $0 \leq n \leq 3$ at the coder module in ADC is shown in **Figure Q2(b)**, determine the source of the encoded signals if the discrete signals are quantized by using up-truncation technique.

(12 marks)

- Q3** (a) Compute the Discrete Fourier Transform (DFT) of $x[n] = (0.9)^n$ for $0 \leq n \leq 3$.
(5 marks)
- (b) Fast Fourier Transform (FFT) is a collection of algorithms for fast computation of DFT. Given the Decimation in Time-FFT (DIT-FFT) of $a[n]$ and $b[n]$ are as follows:

$$A[k] = [-j, 1, -2j, 1]$$
$$B[k] = [0, 2, 0, 2]$$

- (i) Sketch butterfly diagram of the DIT-FFT process for both $A[k]$ and $B[k]$. State the Twiddle factors in each diagram.
(5 marks)
- (ii) Analyze the butterfly diagram in **Q3 (b)(i)** to compute $b[n]$.
(7 marks)
- (iii) Plot the magnitude spectrum of $c[n] = a[n] \otimes b[n]$.
(3 marks)

Q4 (a) Convert the discrete signal below to z-domain, using the defining relation. Then identify the ROC for each of the signal.

(i) $x[n] = \delta[n + 1] + 2\delta[n - 2]$ (2 marks)

(ii) $y[n] = \text{tri}\left(\frac{n-3}{3}\right)$ (2 marks)

(b) The z-transform of $x[n]$ is $X(z) = \frac{4z}{(z + 0.5)^2}$, $|z| > 0.5$. Solve the z-transform of the following using properties:

(i) $y[n] = (2)^n n x[n]$ (3 marks)

(ii) $h[n] = x[n] - x[n - 1]$ (3 marks)

(c) The z-transform of the convolution $y(n) = x(n) * h(n)$ is:

$$Y(z) = \frac{-3z^{-1}}{1 - 3.5z^{-1} + 1.5z^{-2}}$$

Where $x[n] = u[-n]$ and $h[n] = (0.5)^n u[n]$. The ROC of $Y(z)$ is $\frac{1}{2} < |z| < 3$.

(i) Determine the inverse z-transform using partial fraction. (7 marks)

(ii) Find the stability of the system. (3 marks)

- Q5** (a) State **TWO (2)** comparison between the Infinite Impulse Response (IIR) filter and Finite Impulse Response (FIR) filter. (4 marks)
- (b) Explain why the digital filter is better than analog filter. (5 marks)
- (c) A digital low pass filter described by $H(z) = \frac{z + 1}{z^2 - z + 0.2}$ has a cut off frequency, $f = 0.5$ kHz and operates at a sampling frequency, $S = 15$ kHz. By using the given filter, construct a digital low pass filter with a cut off frequency of 2 kHz. (11 marks)

-END OF QUESTIONS -

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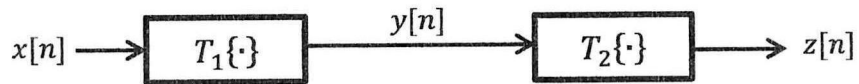


Figure Q1(a)

Encoded signal $x_c[n]$ for $0 \leq n \leq 3$

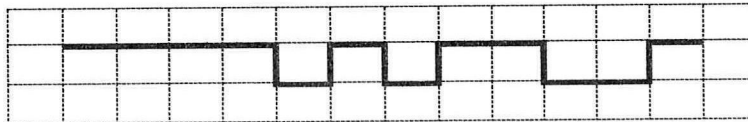


Figure Q2(b)

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Table 1: Properties of the Discrete Fourier Transform (DFT)

Property	Signal	DFT	Remarks
Shift	$x[n - n_o]$	$X_{DFT}[k] e^{-j2\pi k n_o / N}$	No change in magnitude.
Shift	$x[n - 0.5N]$	$(-1)^k X_{DFT}[k]$	Half-period shift for even N .
Modulation	$x[n] e^{j2\pi k_o n / N}$	$X_{DFT}[k - k_o]$	
Modulation	$(-1)^n x[n]$	$X_{DFT}[k - 0.5N]$	Half-period shift for even N .
Folding	$x[-n]$	$X_{DFT}[-k]$	This is circular folding.
Product	$x[n]y[n]$	$\frac{1}{N} X_{DFT}[k] \otimes Y_{DFT}[k]$	The convolution is periodic.
Convolution	$x[n] \otimes y[n]$	$X_{DFT}[k] Y_{DFT}[k]$	The convolution is periodic.
Correlation	$x[n] \otimes \otimes y[n]$	$X_{DFT}[k] Y_{DFT}^*[k]$	The correlation is periodic.
Central Ordinates	$x[0] = \frac{1}{N} \sum_{k=0}^{N-1} X_{DFT}[k], \quad X_{DFT}[0] = \sum_{n=0}^{N-1} x[n]$		
Central Ordinates	$x\left[\frac{N}{2}\right] = \frac{1}{N} \sum_{k=0}^{N-1} (-1)^k X_{DFT}[k] \quad (N \text{ even}),$ $X_{DFT}\left[\frac{N}{2}\right] = \sum_{n=0}^{N-1} (-1)^n x[n] \quad (N \text{ even})$		
Parseval's Relation	$\sum_{n=0}^{N-1} x[n] ^2 = \frac{1}{N} \sum_{k=0}^{N-1} X_{DFT}[k] ^2$		

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Table 2: Properties of the z- transform

Property	Signal	z-transform
Linearity	$a_1x_1[n] + a_2x_2[n]$	$a_1X_1(z) + a_2X_2(z)$
Time reversal	$x[-n]$	$X(z^{-1})$
Time shifting	i) $x(n - k)$ ii) $x(n + k)$	i) $z^{-k}X(z)$ ii) $z^kX(z)$
Convolution	$x_1(n) * x_2(n)$	$X_1(z)X_2(z)$
Correlation	$r_{x_1x_2}(l) = \sum_{n=-\infty}^{\infty} x_1(n)x_2(n-l)$	$R_{x_1x_2}(z) = X_1(z)X_2(z^{-1})$
Scaling	$a^n x(n)$	$X(a^{-1}z)$
Differentiation	$nx[n]$	$z^{-1} \frac{dX(z)}{dz^{-1}}$ or $-z \frac{dX(z)}{dz}$
Time differentiation	$x[n] - x[n-1]$	$X(z)(1 - z^{-1})$
Time integration	$\sum_{k=0}^{\infty} X(k)$	$X(z) = \left(\frac{z}{z-1} \right)$
Initial value theorem	$\lim_{n \rightarrow 0} x(n)$	$\lim_{ z \rightarrow \infty} X(z)$
Final value theorem	$\lim_{n \rightarrow \infty} x(n)$	$\lim_{ z \rightarrow 1} \left(\frac{z-1}{z} \right) X(z)$

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Table 3: z-Transform Pairs

Signal $x(t)$	Sequence $x(n)$	z-Transform $X(z)$
$\delta(t)$	$\delta(n)$	1
$\delta(t - k)$	$\delta(n - k)$	z^{-k}
$u(t)$	$u(n)$	$\frac{1}{1 - z^{-1}} = \frac{z}{z - 1}$
	$-u(-n - 1)$	$\frac{1}{1 - z^{-1}} = \frac{z}{z - 1}$
$r(t) = tu(t)$	$nu(n)$	$\frac{z^{-1}}{(1 - z^{-1})^2} = \frac{z}{(z - 1)^2}$
	$a^n u(n)$	$\frac{1}{1 - az^{-1}} = \frac{z}{z - a}$
	$-a^n u(-n - 1)$	$\frac{1}{1 - az^{-1}} = \frac{z}{z - a}$
	$na^n u(n)$	$\frac{az}{(z - a)^2}$
	$-na^n u(-n - 1)$	$\frac{az}{(z - a)^2}$
e^{-at}	e^{-an}	$\frac{1}{1 - e^{-a} z^{-1}} = \frac{z}{z - e^{-a}}$
t^2	$n^2 u(n)$	$z^{-1} \frac{(1 + z^{-1})}{(1 - z^{-1})^3} = \frac{z(z + 1)}{(z - 1)^3}$
te^{-at}	ne^{-an}	$\frac{z^{-1} e^{-a}}{(1 - e^{-a} z^{-1})^2} = \frac{ze^{-a}}{(z - e^{-a})^2}$
$\sin \omega_0 t$	$\sin \omega_0 n$	$\frac{z \sin \omega_0}{z^2 - 2z \cos \omega_0 + 1}$
$\cos \omega_0 t$	$\cos \omega_0 n$	$\frac{z(z - \cos \omega_0)}{z^2 - 2z \cos \omega_0 + 1}$

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Table 4: Digital- to- digital Transformations

Form	Band Edges	Mapping $s \rightarrow$	Parameters
Lowpass to lowpass	Ω_C	$\frac{z - \alpha}{1 - \alpha z}$	$\alpha = \frac{\sin[0.5(\Omega_D - \Omega_C)]}{\sin[0.5(\Omega_D + \Omega_C)]}$
Lowpass to highpass	Ω_C	$\frac{-(z + \alpha)}{1 + \alpha z}$	$\alpha = \frac{-\cos[0.5(\Omega_D + \Omega_C)]}{\cos[0.5(\Omega_D - \Omega_C)]}$
Lowpass to bandpass	$[\Omega_1, \Omega_2]$	$\frac{-(z^2 + A_1 z + A_2)}{A_2 z^2 + A_1 z + 1}$	$K = \frac{\tan(0.5\Omega_D)}{\tan[0.5(\Omega_2 - \Omega_1)]}$ $\alpha = \frac{-\cos[0.5(\Omega_2 + \Omega_1)]}{\cos[0.5(\Omega_2 - \Omega_1)]}$ $A_1 = \frac{2\alpha K}{K + 1}, A_2 = \frac{K - 1}{K + 1}$
Lowpass to bandstop	$[\Omega_1, \Omega_2]$	$\frac{(z^2 + A_1 z + A_2)}{A_2 z^2 + A_1 z + 1}$	$K = \tan(0.5\Omega_D)\tan[0.5(\Omega_2 - \Omega_1)]$ $\alpha = \frac{-\cos[0.5(\Omega_2 + \Omega_1)]}{\cos[0.5(\Omega_2 - \Omega_1)]}$ $A_1 = \frac{2\alpha}{K + 1}, A_2 = \frac{1 - K}{1 + K}$

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Table 5 : Direct Analog- to- digital Transformations for Bilinear Design

From	Band Edges	Mapping s →	Parameters
Lowpass to lowpass	Ω_c	$\frac{z-1}{C(z+1)}$	$C = \tan(0.5\Omega_c)$
Lowpass to highpass	Ω_c	$\frac{C(z+1)}{z-1}$	$C = \tan(0.5\Omega_c)$
Lowpass to bandpass	$\Omega_1 < \Omega_0 < \Omega_2$	$\frac{z^2 - 2\beta z + 1}{C(z^2 - 1)}$	$C = \tan[0.5(\Omega_2 - \Omega_1)]$, $\beta = \cos\Omega_0$ or $\beta = \frac{\cos[0.5(\Omega_2 + \Omega_1)]}{\cos[0.5(\Omega_2 - \Omega_1)]}$
Lowpass to bandstop	$\Omega_1 < \Omega_0 < \Omega_2$	$\frac{C(z^2 - 1)}{z^2 - 2\beta z + 1}$	$C = \tan[0.5(\Omega_2 - \Omega_1)]$, $\beta = \cos\Omega_0$ or $\beta = \frac{\cos[0.5(\Omega_2 + \Omega_1)]}{\cos[0.5(\Omega_2 - \Omega_1)]}$

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Table 6: Windows for FIR filter design.

Window	Expression $w_N[n]$, $-0.5(N-1) \leq n \leq 0.5(N-1)$
Boxcar	1
Cosine	$\cos\left(\frac{n\pi}{N-1}\right)$
Riemann	$\text{sinc}^L\left(\frac{2n}{N-1}\right), L > 0$
Bartlett	$1 - \frac{2 n }{N-1}$
Von Hann (Hanning)	$0.5 + 0.5\cos\left(\frac{2n\pi}{N-1}\right)$
Hamming	$0.54 + 0.46\cos\left(\frac{2n\pi}{N-1}\right)$
Blackman	$0.42 + 0.5\cos\left(\frac{2n\pi}{N-1}\right) + 0.08\cos\left(\frac{4n\pi}{N-1}\right)$
Kaiser	$\frac{I_0\left(\pi\beta\sqrt{1-4\left(\frac{n}{N-1}\right)^2}\right)}{I_0(\pi\beta)}$

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Table 7 : Characteristics of the windowed spectrum for various windows.

Window	Peak Ripple $\delta_p = \delta_s$	Passband Attenuation $A_{WP}(\text{dB})$	Peak Sidelobe Attenuation $A_{WS}(\text{dB})$	Transition Width $F_{WS} \approx C/N$
Boxcar	0.0897	1.5618	21.7	$C = 0.92$
Cosine	0.0207	0.3600	33.8	$C = 2.10$
Riemann	0.0120	0.2087	38.5	$C = 2.50$
von Hann (Hanning)	0.0063	0.1103	44.0	$C = 3.21$
Hamming	0.0022	0.0384	53.0	$C = 3.47$
Blackman	1.71×10^{-4}	2.97×10^{-3}	75.3	$C = 5.71$

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Identity

$$e^{\pm j\theta} = \cos\theta \pm j\sin\theta$$

$$\cos\theta = \frac{1}{2}(e^{j\theta} + e^{-j\theta})$$

$$\sin\theta = \frac{1}{j2}(e^{j\theta} - e^{-j\theta})$$

Finite Summation Formula

$$\sum_{k=1}^n k = \frac{n(n+1)}{2}$$

$$\sum_{k=1}^n k^2 = \frac{n(n+1)(2n+1)}{6}$$

$$\sum_{k=1}^n k^3 = \frac{n^2(n+1)^2}{4}$$

$$\sum_{k=0}^n \alpha^k = \frac{1-\alpha^{n+1}}{1-\alpha}, \quad \alpha \neq 1$$

$$\sum_{k=0}^n k\alpha^k = \frac{\alpha[1-(n+1)\alpha^n + n\alpha^{n+1}]}{(1-\alpha)^2}$$

$$\sum_{k=0}^n k^2\alpha^k = \frac{\alpha[(1+\alpha)-(n+1)^2\alpha^n + (2n^2+2n-1)\alpha^{n+1} - n^2\alpha^{n+2}]}{(1-\alpha)^3}$$

Infinite Summation Formula

$$\sum_{k=0}^{\infty} \alpha^k = \frac{1}{1-\alpha}, \quad |\alpha| < 1$$

$$\sum_{k=1}^{\infty} \alpha^k = \frac{\alpha}{1-\alpha}, \quad |\alpha| < 1$$

$$\sum_{k=1}^{\infty} k\alpha^k = \frac{\alpha}{(1-\alpha)^2}, \quad |\alpha| < 1$$

$$\sum_{k=1}^{\infty} k^2\alpha^k = \frac{\alpha^2 + \alpha}{(1-\alpha)^3}, \quad |\alpha| < 1$$

$$\sum_{k=-\infty}^{\infty} e^{-\alpha|k|} = \frac{1+e^{-\alpha}}{1-e^{-\alpha}}, \quad \alpha > 0$$

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