



UTHM

Universiti Tun Hussein Onn Malaysia

UNIVERSITI TUN HUSSEIN ONN MALAYSIA

**FINAL EXAMINATION
SEMESTER I
SESSION 2017/2018**

COURSE NAME : DIGITAL SIGNAL PROCESSING
COURSE CODE : BEB 30503
PROGRAMME CODE : BEJ / BEV
EXAMINATION DATE : DECEMBER 2017 / JANUARY 2018
DURATION : 3 HOURS
INSTRUCTION : ANSWER ALL QUESTIONS

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THIS QUESTION PAPER CONSISTS OF **THIRTEEN (13)** PAGES

Q1 (a) Given a system as shown in **Figure Q1(a)** with the impulse response, $h[n] = \{1, 2, 1, 4\}$ and $g[n] = \{1, 3, 5\}$.

(i) Derive a mathematical equation of the given system. (1 mark)

(ii) Analyze the given system for $x[n] = \{1, 2, 3\}$ to confirm that the system can be simplified as:

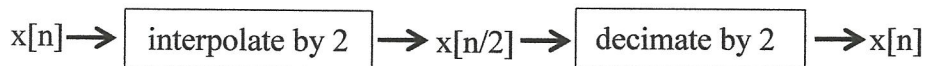
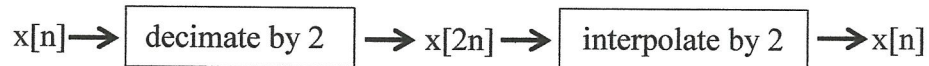
$$z[n] = 3[x[n] * (h[n] + g[n])]$$

(12 marks)

(b) A signal, $x[n] = \{1, -3, 4, -2\}$ is measured by a digital oscilloscope.

(i) Sketch the measured signal. (2 marks)

(ii) Consider two sets of operations as the followings:



Prove that the second set of operations will recovered the $x[n]$ exactly.

(5 marks)

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Q2 (a) With the aid of a diagram, explain the relation between actual and aliased frequency in sampling process.

(3 marks)

(b) A sinusoidal signal, $g(t) = 2.5 \sin(800\pi t)$ V is passed to an analog to digital converter (ADC). The sampling frequency of 20% above the Nyquist Rate is used. It will be encoded with 4 bits per sample using offset binary representation and the dynamic range is ± 3 V. Compare the encoded signal if the sampling frequency of 20% and 25% above the Nyquist Rate using the truncation techniques for $-1 < n < 4$.

(17 marks)

Q3 (a) Elaborate a difference between even and odd conjugate symmetry of a N samples signal where N is an integer value.

(2 marks)

(b) Analyze the following Discrete Fourier Transform (DFT) by proving that there is a suitable property of DFT to relate with $x[n] = \{4, 0, 1\}$ and $X[k] = \{5, 3.5 + j0.866, 3.5 - j0.866\}$.

$$(i) \quad Y_{DFT}[k] = \{5, 3.5 - j0.866, 3.5 + j0.866\}$$

(4 marks)

$$(ii) \quad G_{DFT}[k] = \{17, 15.5 + j0.866, 15.5 - j0.866\}$$

(4 marks)

Hints :

$$e^{j\frac{2\pi}{3}} = e^{j\frac{8\pi}{3}} = -0.5 + j0.866$$

$$e^{j\frac{4\pi}{3}} = -0.5 - j0.866$$

(c) **Figure Q3(c)** shows twiddle factors and values at intermediate nodes of 4-point ABC algorithm.

(i) State the name of ABC algorithm.

(2 marks)

(ii) Compute the discrete signal $c[n]$ and the DFT signal $C_{DFT}[k]$.

(8 marks)

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- Q4 (a)** Based on **Figure Q1(a)**, calculate the signal, $z[n]$ by applying the z-transform approach. The simplification of the system is not required. All the value of $h[n]$, $g[n]$ and $x[n]$ are based on question **Q1**.

(12 marks)

- (b) Determine the inverse z-transform of $X(z) = \frac{z^3}{(z-0.3)(z-0.5)^2}$ using the partial fraction expansion method.

Hints:

$$H(z) = \frac{N(z)}{(z-p_1)(z-p_2)\dots(z-p_n)} = \frac{A_1}{(z-p_1)} + \frac{A_2}{(z-p_2)} + \dots + \frac{A_n}{(z-p_n)}$$

$$A_i = (z-p_i)H(z)|_{z=p_i}$$

$$\frac{A_{i1}}{(z-p_i)} + \frac{A_{i2}}{(z-p_i)^2} + \dots + \frac{A_{ir}}{(z-p_i)^r}$$

$$A_{ir} = (z-p_i)^r H(z)|_{z=p_i}$$

$$A_{i(r-1)} = \frac{d}{dz} (z-p_i)^r H(z)|_{z=p_i}$$

$$A_{i(r-k)} = \frac{1}{k!} \frac{d^k}{dz^k} (z-p_i)^r H(z)|_{z=p_i}$$



(8 marks)

- Q5 (a)** Design a finite impulse response (FIR) highpass filter having a cut-off frequency of 5 kHz with the sample length of 9 for operating at the sampling frequency of 20 kHz using cosinus window.

(13 marks)

- (b) Given a differential equation of a filter as:

$$y[n] = 0.5x[n] + 0.5x[n+2] - y[n-1]$$

- (i) Prove that this differential equation is referring to an IIR filter

(4 marks)

- (ii) Analyze this filter system using an input signal of $x[n] = \{2, 3, -2\}$ and $y[-1] = 0$.

(3 marks)

-END OF QUESTIONS -

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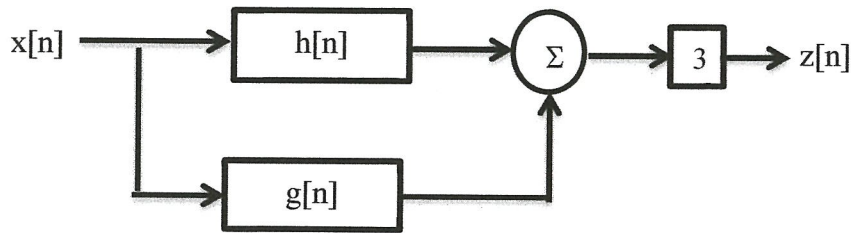


Figure Q1(a)

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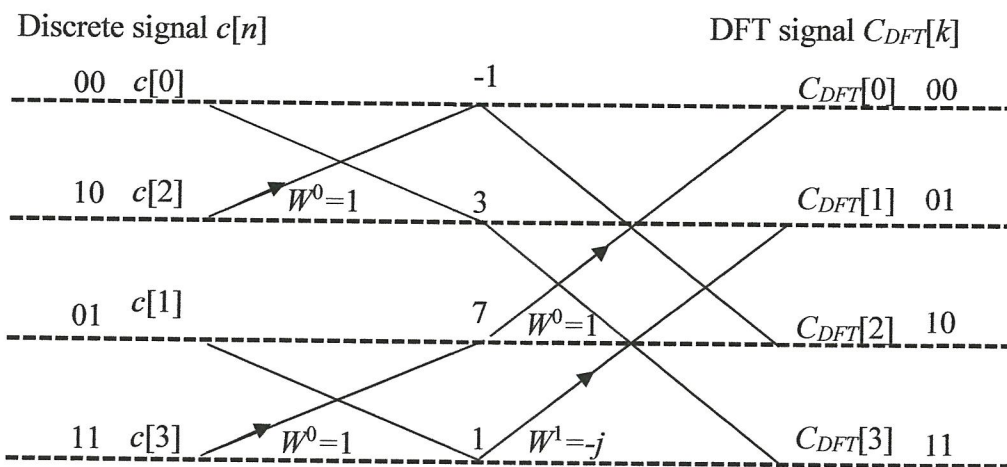


Figure Q3(c)

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

Property	Signal	DFT	Remarks
Shift	$x[n - n_o]$	$X_{DFT}[k]e^{-j2\pi kn_o/N}$	No change in magnitude.
Shift	$x[n - 0.5N]$	$(-1)^k X_{DFT}[k]$	Half-period shift for even N .
Modulation	$x[n]e^{j2\pi nk_o/N}$	$X_{DFT}[k - k_o]$	
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	$r_{x_1x_2}(l) = \sum_{n=-\infty}^{\infty} x_1(n)x_2(n-l)$	$R_{x_1x_2}(z) = X_1(z)X_2(z^{-l})$
Scaling	$a^n x(n)$	$X(az^{-1})$
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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