

**University of Eswatini**  
**Faculty of Science and Engineering**  
**Department of Electrical and Electronic Engineering**  
**Supplementary Exam 2019**

---

**Title of Paper** : Introduction to Digital Signal Processing

**Course Number** : EEE446/EE443

---

**Time Allowed** : 3 hrs

**Instructions** :

1. Answer all the questions
2. Each question carries 25 marks
3. Useful information is attached at the end of the question paper
4. Make sure there are 6 pages including the cover page

**THIS PAPER SHOULD NOT BE OPENED UNTIL PERMISSION HAS  
BEEN GIVEN BY THE INVIGILATOR**

### QUESTION 1

- (a) A digital signal processing (DSP) system is described by the difference equation

$$y(n) + 0.6y(n-1) - 0.4y(n-2) = x(n) + x(n-1)$$

Determine the solution when the initial conditions are zero and  $x(n) = u(n)$ .

(10 marks)

- (b) The impulse response  $h(n)$ , and input  $x(n)$  are given as  $x(n) = u(n) - u(n-2)$ ;  $h(n) = \left(\frac{1}{3}\right)^n, n \geq 0$ . Find the response  $y(n)$  in closed form.

(10 marks)

- (c) Find the z-transform of  $x(n) = \left(\frac{1}{2}\right)^n [u(n) - u(n-10)]$ .

(5 marks)

### QUESTION 2

- (a) Define circular convolution. For the sequences  $x_1(n) = \{1, 2, 3, 1\}$  and  $x_2(n) = \{4, 3, 2, 2\}$ ,

Determine the 4-point circular convolution of  $x_3(n) = x_1(n) \otimes x_2(n)$ .

(10 marks)

- (b) What is the advantage of using fast Fourier transform (FFT) algorithm to determine the DFT of a discrete sequence? Using DIT-FFT algorithm, determine the 8-point DFT of the sequence  $x(n) = \{1, 1, 0, 1, 1, 0, 0, 1\}$ . What is the speed improvement factor in this case?

(15 marks)

### QUESTION 3

- (a) Design a 5-tap FIR band reject (band-stop) filter with a lower cut-off frequency of 2,000 Hz, an upper cut-off frequency of 2,400 Hz, and a sampling rate of 8,000 Hz using the Hamming window method. Determine the transfer function and difference equation of the desired FIR system.

(20 marks)

- (b) Determine the causal signal  $x(n)$  having the z-transform

$$X(z) = \frac{1}{(1+z^{-1})(1-z^{-1})^2} \quad (5 \text{ marks})$$

#### QUESTION 4

- (a) Design a first-order high-pass digital Chebyshev filter with a cut-off frequency of 3 kHz and 1 dB ripple on the pass-band using a sampling frequency of 8,000 Hz. (Use Bilinear Transformation method.)

State the difference equation and the transfer function of the filter designed

(15 marks)

- (b) Use direct Form I and II to realize the filter.

(10 marks)

**Table 1: Properties of z-transform**

Property	Time Domain	z-Transform
Linearity	$ax_1(n) + bx_2(n)$	$aZ(x_1(n)) + bZ(x_2(n))$
Shift theorem	$x(n - m)$	$z^{-m}X(z)$
Linear convolution	$x_1(n) * x_2(n) = \sum_{k=0}^n x_1(n - k)x_2(k)$	$X_1(z)X_2(z)$

**Differentiation in the z-domain**       $nx(n)$        $-z \frac{dX(z)}{dz}$

**TABLE 8.5 Chebyshev lowpass prototype transfer functions with 1 dB ripple ( $r = 0.5088$ )**

$n$	$H_p(s)$
1	$\frac{1.9652}{s+1.9652}$
2	$\frac{0.9826}{s^2+1.0977s+1.1025}$
3	$\frac{0.4913}{s^3+0.9883s^2+1.2384s+0.4913}$
4	$\frac{0.2456}{s^4-0.9528s^3+1.4539s^2+0.7426s+0.2756}$
5	$\frac{0.1228}{s^5+0.9368s^4+1.6888s^3+0.9744s^2+0.5805s+0.1228}$
6	$\frac{0.0614}{s^6-0.9283s^5+1.9308s^4+1.20121s^3+0.9393s^2+0.3071s+0.0689}$

**Table 3: Summary of ideal impulse responses for standard FIR filters.**

Filter Type	Ideal Impulse Response $h(n)$ (noncausal FIR coefficients)
Lowpass:	$h(n) = \begin{cases} \frac{\Omega_c}{\pi} & n = 0 \\ \frac{\sin(\Omega_c n)}{n\pi} & \text{for } n \neq 0 \end{cases} \quad M \leq n \leq M$
Highpass:	$h(n) = \begin{cases} \frac{\pi - \Omega_c}{\pi} & n = 0 \\ -\frac{\sin(\Omega_c n)}{n\pi} & \text{for } n \neq 0 \end{cases} \quad -M \leq n \leq M$
Bandpass:	$h(n) = \begin{cases} \frac{\Omega_H - \Omega_L}{\pi} & n = 0 \\ \frac{\sin(\Omega_H n)}{n\pi} - \frac{\sin(\Omega_L n)}{n\pi} & \text{for } n \neq 0 \end{cases} \quad -M \leq n \leq M$
Bandstop:	$h(n) = \begin{cases} \frac{\pi - \Omega_H + \Omega_L}{\pi} & n = 0 \\ -\frac{\sin(\Omega_H n)}{n\pi} + \frac{\sin(\Omega_L n)}{n\pi} & \text{for } n \neq 0 \end{cases} \quad -M \leq n \leq M$

Causal FIR filter coefficients: shifting  $h(n)$  to the right by  $M$  samples.

Transfer function:

$$H(z) = b_0 + b_1z^{-1} + b_2z^{-2} + \dots + b_{2M}z^{-2M}$$

where  $b_n = h(n - M)$ ,  $n = 0, 1, \dots, 2M$

**Table 4:** FIR filter length estimation using window functions (normalized transition width  $\Delta f = |f_{stop} - f_{pass}|/f_s$ ).

Window Type	Window Function $w(n), -M \leq n \leq M$	Window Length, $N$	Passband Ripple (dB)	Stopband Attenuation (dB)
Rectangular	1	$N = 0.9/\Delta f$	0.7416	21
Hanning	$0.5 + 0.5 \cos\left(\frac{\pi n}{M}\right)$	$N = 3.1/\Delta f$	0.0546	44
Hamming	$0.54 + 0.46 \cos\left(\frac{\pi n}{M}\right)$	$N = 3.3/\Delta f$	0.0194	53
Blackman	$0.42 + 0.5 \cos\left(\frac{\pi n}{M}\right) + 0.08 \cos\left(\frac{2\pi n}{M}\right)$	$N = 5.5/\Delta f$	0.0017	74

**Table 5:** Analog lowpass prototype transformations

Filter Type	Prototype Transformation
Lowpass	$\frac{s}{\omega_c}, \omega_c$ is the cutoff frequency
Highpass	$\frac{\omega_c}{s}, \omega_c$ is the cutoff frequency
Bandpass	$\frac{s^2 + \omega_0^2}{sW}, \omega_0 = \sqrt{\omega_l \omega_h}, W = \omega_h - \omega_l$
Bandstop	$\frac{sW}{s^2 + \omega_0^2}, \omega_0 = \sqrt{\omega_l \omega_h}, W = \omega_h - \omega_l$

**Table 6:** Conversion from analog filter specifications to lowpass prototype specifications.

Analog Filter Specifications	Lowpass Prototype Specifications
Lowpass: $\omega_{ap}, \omega_{as}$	$v_p = 1, v_s = \omega_{as}/\omega_{ap}$
Highpass: $\omega_{ap}, \omega_{as}$	$v_p = 1, v_s = \omega_{ap}/\omega_{as}$
Bandpass: $\omega_{apl}, \omega_{aph}, \omega_{ast}, \omega_{ash}$ $\omega_0 = \sqrt{\omega_{apl}\omega_{aph}}, \omega_0 = \sqrt{\omega_{ast}\omega_{ash}}$	$v_p = 1, v_s = \frac{\omega_{ash} - \omega_{ast}}{\omega_{aph} - \omega_{apl}}$
Bandstop: $\omega_{apl}, \omega_{aph}, \omega_{ast}, \omega_{ash}$ $\omega_0 = \sqrt{\omega_{apl}\omega_{aph}}, \omega_0 = \sqrt{\omega_{ast}\omega_{ash}}$	$v_p = 1, v_s = \frac{\omega_{aph} - \omega_{apl}}{\omega_{ast} - \omega_{ash}}$

$\omega_{ap}$ , passband frequency edge;  $\omega_{as}$ , stopband frequency edge;  $\omega_{apl}$ , lower cutoff frequency in passband;  $\omega_{aph}$ , upper cutoff frequency in passband;  $\omega_{ast}$ , lower cutoff frequency in stopband;  $\omega_{ash}$ , upper cutoff frequency in stopband;  $\omega_0$ , geometric center frequency.

## The Z-transform

Line No.	$x(n), n \geq 0$	z-Transform $X(z)$	Region of Convergence
1	$x(n)$	$\sum_{n=0}^{\infty} x(n)z^{-n}$	
2	$\delta(n)$	1	$ z  > 0$
3	$au(n)$	$\frac{az}{z-a}$	$ z  > a$
4	$nu(n)$	$\frac{z}{(z-1)^2}$	$ z  > 1$
5	$n^2u(n)$	$\frac{z(z+1)}{(z-1)^3}$	$ z  > 1$
6	$a^n u(n)$	$\frac{z}{z-a}$	$ z  > a$
7	$e^{-an} u(n)$	$\frac{z}{z-e^{-a}}$	$ z  > e^{-a}$
8	$na^n u(n)$	$\frac{az}{(z-a)^2}$	$ z  > a$
9	$\sin(an)u(n)$	$\frac{z \sin(a)}{z^2 - 2z \cos(a) + 1}$	$ z  > 1$
10	$\cos(an)u(n)$	$\frac{z[z - \cos(a)]}{z^2 - 2z \cos(a) + 1}$	$ z  > 1$
11	$a^n \sin(bn)u(n)$	$\frac{[a \sin(b)]z}{z^2 - [2a \cos(b)]z + a^2}$	$ z  > a$
12	$a^n \cos(bn)u(n)$	$\frac{z[z - a \cos(b)]}{z^2 - [2a \cos(b)]z + a^2}$	$ z  > a$
13	$e^{-an} \sin(bn)u(n)$	$\frac{[e^{-a} \sin(b)]z}{z^2 - [2e^{-a} \cos(b)]z + e^{-2a}}$	$ z  > e^{-a}$
14	$e^{-an} \cos(bn)u(n)$	$\frac{z[z - e^{-a} \cos(b)]}{z^2 - [2e^{-a} \cos(b)]z + e^{-2a}}$	$ z  > e^{-a}$
15	$2A/P^n \cos(n\theta + \phi)u(n)$ where $P$ and $A$ are complex constants defined by $P = P e^{-j\theta}, A =  A  e^{j\phi}$	$\frac{Az}{z-P} + \frac{A^*z}{z-P^*}$	