

NATIONAL UNIVERSITY OF SINGAPORE

EXAMINATION FOR
(Semester I : 2014/2015)

EE2023 – SIGNALS & SYSTEMS

Nov/Dec 2014 - Time Allowed: 2.5 Hours

INSTRUCTIONS TO CANDIDATES

1. This paper contains **EIGHT (8)** questions and comprises **ELEVEN (11)** printed pages.
2. Answer **ALL** questions in **Section A** and **ANY THREE (3)** questions in **Section B**.
3. This is a **CLOSED BOOK** examination.
4. Programmable calculators are not allowed.
5. Tables of Fourier Transforms, Laplace Transforms and trigonometric identities are provided in Pages 9, 10 and 11, respectively.

SECTION A : Answer ALL questions in this section

Q1. Consider the system in Figure Q1-1 whose transfer function is given by $G(s) = \frac{1}{s} e^{-3s}$.

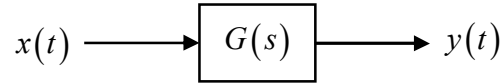


Figure Q1-1 : System, $G(s)$

- (a) Derive the response of $G(s)$ to a unit impulse, $x(t) = \delta(t)$. Sketch the resulting impulse response.

(4 marks)

- (b) Derive and sketch the response of $G(s)$ to an input, $x(t)$ given in Figure Q1-2. Label your sketch clearly.

(6 marks)

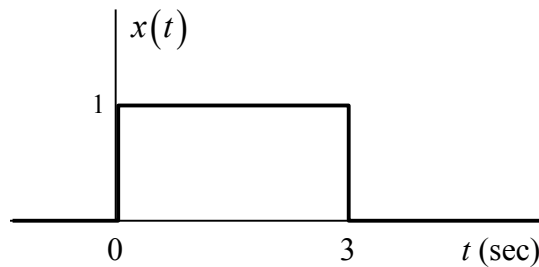


Figure Q1-2 : Input to $G(s)$

Q2. The signal $x(t) = \text{sinc}^2(2t)$ is sampled at 8 Hz to produce the sampled signal $x_s(t)$.

- (a) Derive the expression for $x_s(t)$.

(3 marks)

- (b) Derive the Fourier transform of $x_s(t)$.

(3 marks)

- (c) Sketch the spectrum of $x_s(t)$.

(4 marks)

Q3. An energy signal $x(t)$ is given by

$$x(t) = \exp(-\pi t^2).$$

- (a) Determine the energy spectral density, $E_x(f)$, of $x(t)$. (4 marks)
- (b) Find the 3 dB bandwidth of $x(t)$. (3 marks)
- (c) In computing the total energy of $x(t)$, would the time-domain or frequency-domain approach lead to a simpler solution, and why? (3 marks)

Q4. The pole-zero diagrams for 2 systems (I and II) are shown in Figure Q4.

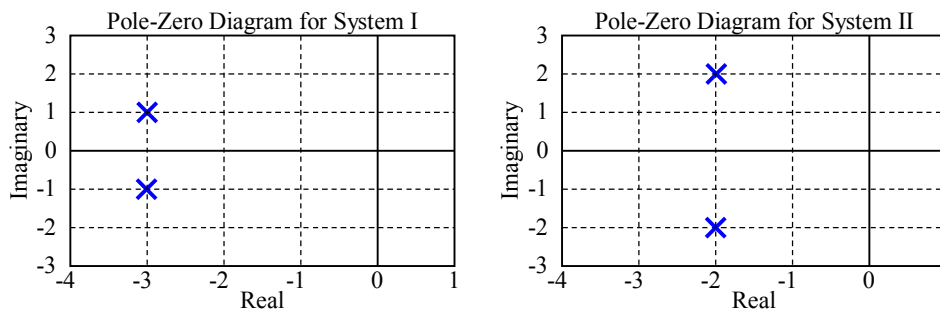


Figure Q4 : Pole-zero diagrams

- (a) What is the damped natural frequency, undamped natural frequency and damping ratio of System I? (3 marks)
- (b) Following a step change of magnitude 3, the steady state output of System II is 15. Derive the transfer function of System II using the pole-zero diagram and the steady-state information. (4 marks)
- (c) Suppose Systems I and II have the same DC gain. Will System I or System II exhibit a larger overshoot following a step change in the input signal? Justify your answer.

Hint : Overshoot is governed by the damping ratio of the system.

(3 marks)

SECTION B : Answer 3 out of the 4 questions in this section

Q5. The space booster in Figure Q5 has a transfer function, $G(s)$, given by

$$\frac{\Phi(s)}{F(s)} = G(s) = \frac{1}{s^2 - 0.04}$$

where $\Phi(s)$ and $F(s)$ are Laplace transforms of $\phi(t)$ and $f(t)$, respectively.

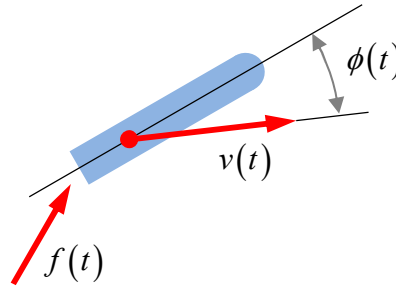


Figure Q5 : A Space Booster

- (a) Analyze and describe what happens to the space booster when it is fired up with $f(t) = \delta(t)$, where $\delta(t)$ is a unit impulse function. Sketch the impulse response, $\phi(t)$.

(4 marks)

- (b) A control system is then developed for the space booster such that the overall closed loop control system has a transfer function given by

$$G_{cl}(s) = \frac{K}{s^2 + K_D s + K_p - 0.04}.$$

- i. If $K_D = 0$, what is the minimum value of K_p required for the closed loop system to have bounded outputs? Justify your answer.

(4 marks)

- ii. If $K_p = 0$, why is K_D alone not able to stabilize the closed loop system?

(4 marks)

- iii. Design K_p and K_D so that the closed loop system has poles at $s_{1,2} = -0.2 \pm 0.3j$.

(4 marks)

- iv. Find the damping ratio and the undamped natural frequency of the closed loop system with poles at $s_{1,2} = -0.2 \pm 0.3j$. Sketch the impulse response of this closed loop system.

(4 marks)

Q.6 Consider the periodic signal $x(t)$ shown in Figure Q6 which comprises periodic Gaussian pulses, where

$$x(t) = \sum_{k=-\infty}^{\infty} e^{-(t-4k)^2/0.25}.$$

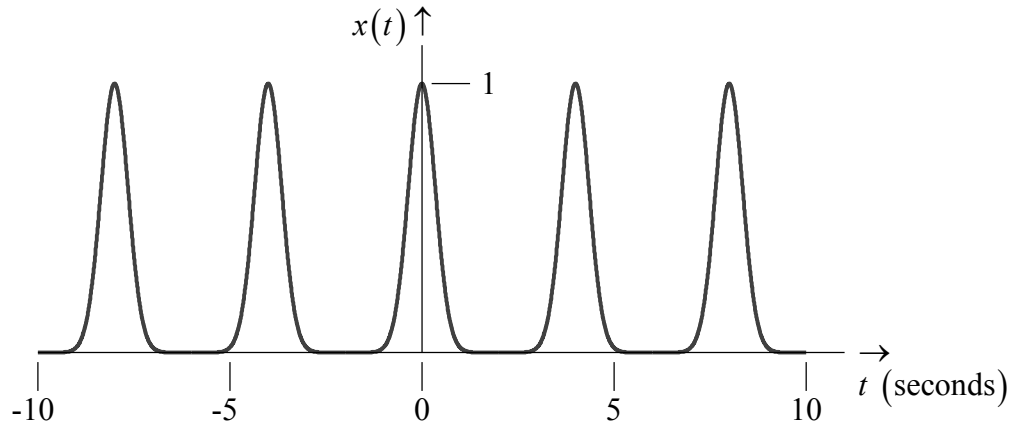


Figure Q6

- (a) Derive the Fourier transform, $X(f)$, of $x(t)$.
(7 marks)
- (b) Derive the Fourier series coefficient, X_k , of $x(t)$.
(3 marks)
- (c) Derive an expression for the average power of $x(t)$.
(3 marks)
- (d) Let the M^{th} harmonic of $x(t)$ be the harmonic that is closest to the 98% containment bandwidth of $x(t)$. Show and explain how M should be found.
(7 marks)

- Q7. Radio station X transmits the signal $x(t) = 10 \cdot \text{sinc}(10t) \cdot \cos(2000\pi t)$, and radio station Y transmits the signal

$$y(t) = m(t) \cdot \cos(2\pi f_c t)$$

where the spectrum of $m(t)$ is given by $M(f) = \text{tri}\left(\frac{f}{B}\right)$, and $f_c \gg 2B$. Radio interference between the two stations will occur if the spectra of their transmissions overlap.

- (a) Sketch of the spectrum, $X(f)$, of $x(t)$. Show all the important dimensions in your sketch. (5 marks)
- (b) Suppose $B = 8$. What is the range of f_c values that should be avoided by radio station Y so as to avoid radio interference between the two stations? (5 marks)
- (c) Suppose $f_c = 1020$. Find the maximum value of B that can be used by radio station Y without causing radio interference between the two stations. (5 marks)
- (d) Suppose $y(t)$ is sampled to form $y_s(t)$. Suggest a sampling frequency, f_s , so that $m(t)$ can be recovered without distortion by passing $y_s(t)$ through a suitably designed ideal lowpass filter. Explain your answer. (5 marks)

Q8. Most loudspeakers are not capable of reproducing the entire audio spectrum with negligible distortion. Consequently, most hi-fi speaker systems use a combination of loudspeakers, each catering to a different frequency band. Filter systems are used to split the audio signal into frequency bands that can be separately routed to loudspeakers optimized for those bands i.e. a lowpass filter is used to isolate signals for the woofer loudspeaker and the output signal of a highpass filter drives the tweeter loudspeaker.

- (a) Derive the transfer functions of the two filter circuits shown in Figure Q8. (8 marks)

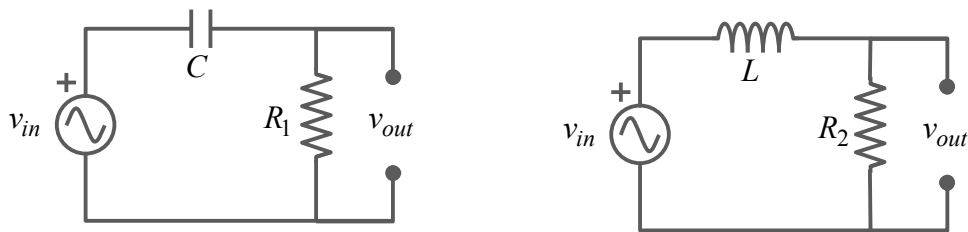


Figure Q8 : Series resistor-capacitor and resistor-inductor filtering circuits

- (b) Sketch the straight line asymptotic Bode Magnitude diagrams of the resistor-capacitor (R_1C) and resistor-inductor (R_2L) filtering circuits, clearly labelling the corner frequencies and the slope of the asymptotes. Hence, or otherwise, determine if the output signal of the R_1C circuit should be used to drive the woofer or the tweeter loudspeaker?

(5 marks)

- (c) A bandpass filtering system for generating audio signals in the mid-frequency range may be constructed by cascading the R_1C and R_2L filtering circuits shown in Figure Q8, and ensuring that $R_1C > \frac{L}{R_2}$.

- i. Sketch the straight line asymptotic Bode Magnitude diagram of the bandpass filter, clearly labelling the corner frequencies.

(3 marks)

- ii. Suppose $R_1 = R_2 = 8\Omega$ and the bandpass system should not distort signals between 800 Hz and 3000 Hz. Design suitable values for C and L .

(4 marks)

END OF QUESTIONS

This page is intentionally left blank to facilitate detachment of the formula sheet for easy reference.

Fourier Series:
$$\begin{cases} X_k = \frac{1}{T} \int_{\tilde{t}}^{\tilde{t}+T} x(t) \exp(-j2\pi k t/T) dt \\ x(t) = \sum_{k=-\infty}^{\infty} X_k \exp(j2\pi k t/T) \end{cases}$$

Fourier Transform:
$$\begin{cases} X(f) = \int_{-\infty}^{\infty} x(t) \exp(-j2\pi f t) dt \\ x(t) = \int_{-\infty}^{\infty} X(f) \exp(j2\pi f t) df \end{cases}$$

FOURIER TRANSFORMS OF BASIC FUNCTIONS		
	$x(t)$	$X(f)$
Constant	K	$K\delta(f)$
Unit Impulse	$\delta(t)$	1
Unit Step	$u(t)$	$\frac{1}{2} \left[\delta(f) + \frac{1}{j\pi f} \right]$
Sign (or Signum)	$\text{sgn}(t)$	$\frac{1}{j\pi f}$
Rectangle	$\text{rect}\left(\frac{t}{T}\right)$	$T \text{sinc}(fT)$
Triangle	$\text{tri}\left(\frac{t}{T}\right)$	$T \text{sinc}^2(fT)$
Sine Cardinal	$\text{sinc}\left(\frac{t}{T}\right)$	$T \text{rect}(fT)$
Complex Exponential	$\exp(j2\pi f_o t)$	$\delta(f - f_o)$
Cosine	$\cos(2\pi f_o t)$	$\frac{1}{2} [\delta(f - f_o) + \delta(f + f_o)]$
Sine	$\sin(2\pi f_o t)$	$-\frac{j}{2} [\delta(f - f_o) - \delta(f + f_o)]$
Gaussian	$\exp\left(-\frac{t^2}{\alpha^2}\right)$	$\alpha \pi^{0.5} \exp(-\alpha^2 \pi^2 f^2)$
Comb	$\sum_{m=-\infty}^{\infty} \delta(t - mT)$	$\frac{1}{T} \sum_{k=-\infty}^{\infty} \delta\left(f - \frac{k}{T}\right)$

FOURIER TRANSFORM PROPERTIES		
	Time-domain	Frequency-domain
Linearity	$\alpha x_1(t) + \beta x_2(t)$	$\alpha X_1(f) + \beta X_2(f)$
Time scaling	$x(\beta t)$	$\frac{1}{ \beta } X\left(\frac{f}{\beta}\right)$
Duality	$X(t)$	$x(-f)$
Time shifting	$x(t - t_o)$	$X(f) \exp(-j2\pi f t_o)$
Frequency shifting (Modulation)	$x(t) \exp(j2\pi f_o t)$	$X(f - f_o)$
Differentiation in the time-domain	$\frac{d^n}{dt^n} x(t)$	$(j2\pi f)^n X(f)$
Multiplication in the time-domain	$x_1(t) x_2(t)$	$\int_{-\infty}^{\infty} X_1(\zeta) X_2(f - \zeta) d\zeta$ or $X_1(f) * X_2(f)$
Convolution in the time-domain	$\int_{-\infty}^{\infty} x_1(\zeta) x_2(t - \zeta) d\zeta$ or $x_1(t) * x_2(t)$	$X_1(f) X_2(f)$
Integration in the time-domain	$\int_{-\infty}^t x(\tau) d\tau$	$\frac{1}{j2\pi f} X(f) + \frac{1}{2} X(0) \delta(f)$
		$\frac{1}{j2\pi f} X(f) \text{ if } X(0) = 0$

$$\text{Unilateral Laplace Transform: } X(s) = \int_{0^-}^{\infty} x(t) \exp(-st) dt$$

LAPLACE TRANSFORMS OF BASIC FUNCTIONS		
	$x(t)$	$X(s)$
Unit Impulse	$\delta(t)$	1
Unit Step	$u(t)$	$1/s$
Ramp	$tu(t)$	$1/s^2$
n th order Ramp	$t^n u(t)$	$\frac{n!}{s^{n+1}}$
Damped Ramp	$t \exp(-\alpha t) u(t)$	$1/(s + \alpha)^2$
Exponential	$\exp(-\alpha t) u(t)$	$1/(s + \alpha)$
Cosine	$\cos(\omega_o t) u(t)$	$s/(s^2 + \omega_o^2)$
Sine	$\sin(\omega_o t) u(t)$	$\omega_o/(s^2 + \omega_o^2)$
Damped Cosine	$\exp(-\alpha t) \cos(\omega_o t) u(t)$	$\frac{s + \alpha}{(s + \alpha)^2 + \omega_o^2}$
Damped Sine	$\exp(-\alpha t) \sin(\omega_o t) u(t)$	$\frac{\omega_o}{(s + \alpha)^2 + \omega_o^2}$

LAPLACE TRANSFORM PROPERTIES		
	Time-domain	s-domain
Linearity	$\alpha x_1(t) + \beta x_2(t)$	$\alpha X_1(s) + \beta X_2(s)$
Time shifting	$x(t - t_o) u(t - t_o)$	$\exp(-st_o) X(s)$
Shifting in the s-domain	$\exp(s_o t) x(t)$	$X(s - s_o)$
Time scaling	$x(\alpha t)$	$\frac{1}{ \alpha } X\left(\frac{s}{\alpha}\right)$
Integration in the time-domain	$\int_{0^-}^t x(\zeta) d\zeta$	$\frac{1}{s} X(s)$
Differentiation in the time-domain	$\frac{dx(t)}{dt}$	$sX(s) - x(0^-)$
	$\frac{d^n x(t)}{dt^n}$	$s^n X(s) - \sum_{k=0}^{n-1} s^{n-1-k} \frac{d^k x(t)}{dt^k} \Big _{t=0^-}$
Differentiation in the s-domain	$-tx(t)$	$\frac{dX(s)}{ds}$
	$(-t)^n x(t)$	$\frac{d^n X(s)}{ds^n}$
Convolution in the time-domain	$\int_{-\infty}^{\infty} x_1(\zeta) x_2(t - \zeta) d\zeta$	$X_1(s) X_2(s)$
Initial value theorem	$x(0^+) = \lim_{s \rightarrow \infty} sX(s)$	
Final value theorem	$\lim_{t \rightarrow \infty} x(t) = \lim_{s \rightarrow 0} sX(s)$	

	$y_{step}(t)$	$Y_{step}(s)$	SYSTEM PARAMETERS
Step response of 1 st order system	$K \left[1 - \exp\left(-\frac{t}{T}\right) \right] u(t)$	$\frac{1}{s} \cdot \frac{K}{(sT+1)}$	$\left(\begin{array}{l} T: \text{System Time-constant} \\ K: \text{System Steady-state (or DC) Gain} \end{array} \right)$
Step response of 2 nd order <u>underdamped</u> system: ($0 < \zeta < 1$)	$K \left[1 - \frac{\exp(-\omega_n \zeta t)}{(1-\zeta^2)^{0.5}} \sin\left(\omega_n (1-\zeta^2)^{0.5} t + \phi\right) \right] u(t)$ $K \left[1 - \left(\frac{\sigma^2 + \omega_d^2}{\omega_d^2} \right)^{0.5} \exp(-\sigma t) \sin(\omega_d t + \phi) \right] u(t)$	$\frac{1}{s} \cdot \frac{K \omega_n^2}{s^2 + 2\zeta \omega_n s + \omega_n^2}$ $\frac{1}{s} \cdot \frac{K(\sigma^2 + \omega_d^2)}{(s + \sigma)^2 + \omega_d^2}$	$\left(\begin{array}{l} \omega_n: \text{System Undamped Natural Frequency} \\ \zeta: \text{System Damping Factor} \\ \omega_d: \text{System Damped Natural Frequency} \\ K: \text{System Steady-state (or DC) Gain} \end{array} \right) \left(\begin{array}{l} \sigma = \omega_n \zeta \\ \omega_d^2 = \omega_n^2 (1 - \zeta^2) \\ \omega_n^2 = \sigma^2 + \omega_d^2 \\ \tan(\phi) = \omega_d / \sigma \end{array} \right)$
2 nd order system - RESONANCE - ($0 \leq \zeta < 1/\sqrt{2}$)	$\text{RESONANCE FREQUENCY: } \omega_r = \omega_n (1 - 2\zeta^2)^{0.5}$		$\text{RESONANCE PEAK: } M_r = \left H(j\omega_r) \right = \frac{K}{2\zeta(1 - \zeta^2)^{0.5}}$

TRIGONOMETRIC IDENTITIES	
$\exp(\pm j\theta) = \cos(\theta) \pm j \sin(\theta)$	$\sin(\alpha \pm \beta) = \sin(\alpha) \cos(\beta) \pm \cos(\alpha) \sin(\beta)$
$\cos(\theta) = \frac{1}{2} [\exp(j\theta) + \exp(-j\theta)]$	$\cos(\alpha \pm \beta) = \cos(\alpha) \cos(\beta) \mp \sin(\alpha) \sin(\beta)$
$\sin(\theta) = \frac{1}{j2} [\exp(j\theta) - \exp(-j\theta)]$	$\tan(\alpha \pm \beta) = \frac{\tan(\alpha) \pm \tan(\beta)}{1 \mp \tan(\alpha) \tan(\beta)}$
$\sin^2(\theta) + \cos^2(\theta) = 1$	
$\sin(2\theta) = 2 \sin(\theta) \cos(\theta)$	$\sin(\alpha) \sin(\beta) = \frac{1}{2} [\cos(\alpha - \beta) - \cos(\alpha + \beta)]$
$\cos(2\theta) = \cos^2(\theta) - \sin^2(\theta)$	$\cos(\alpha) \cos(\beta) = \frac{1}{2} [\cos(\alpha - \beta) + \cos(\alpha + \beta)]$
$\sin^2(\theta) = \frac{1}{2} [1 - \cos(2\theta)]$	$\sin(\alpha) \cos(\beta) = \frac{1}{2} [\sin(\alpha - \beta) + \sin(\alpha + \beta)]$
$\cos^2(\theta) = \frac{1}{2} [1 + \cos(2\theta)]$	$C \cos(\theta) - S \sin(\theta) = \sqrt{C^2 + S^2} \cos\left[\theta + \tan^{-1}\left(\frac{S}{C}\right)\right]$