

**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 whose transfer function is given by  $G(s) = \frac{1}{s}e^{-3s}$ .

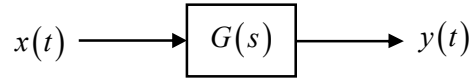


Figure Q1 : System,  $G(s)$

- (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. Label your sketch clearly.

(6 marks)

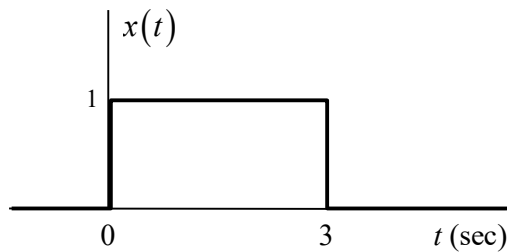


Figure Q1 : Input to  $G(s)$

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. A 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.

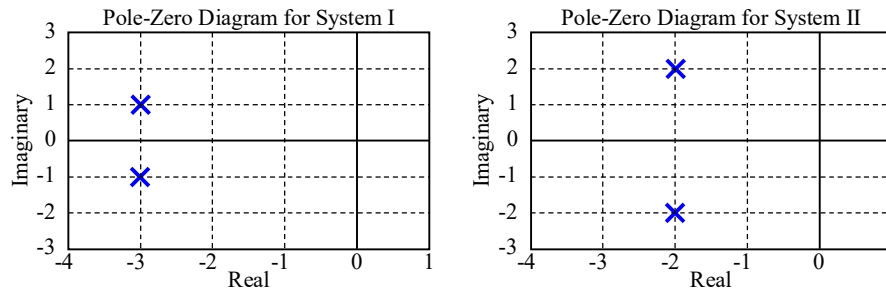


Figure Q4 : Pole-zero diagrams

- (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. (3 marks)
- Hint : Overshoot is governed by the damping ratio of the system.*

## 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.

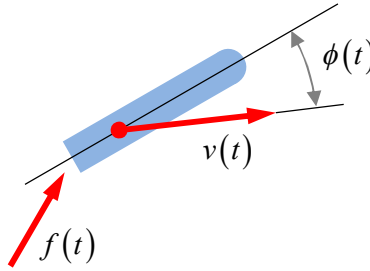


Figure Q5 : A Space Booster

- (a) 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

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

- 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

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

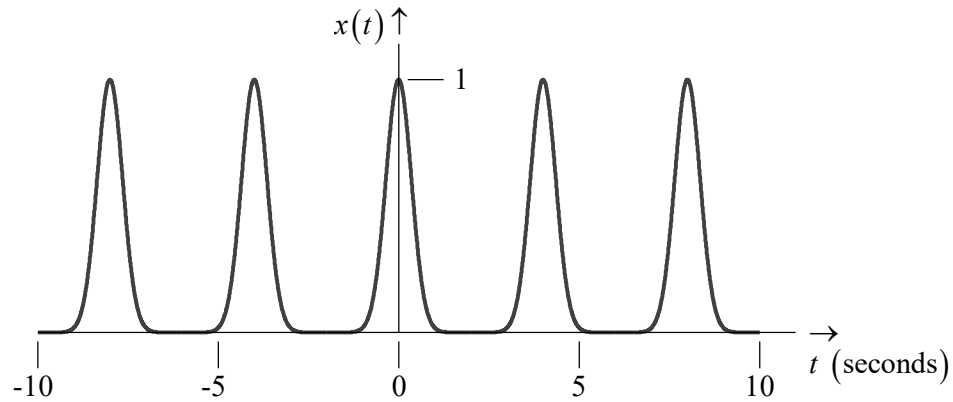


Figure Q6

- (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% power containment bandwidth of  $x(t)$ . Explain how  $M$  could 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 covering 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)

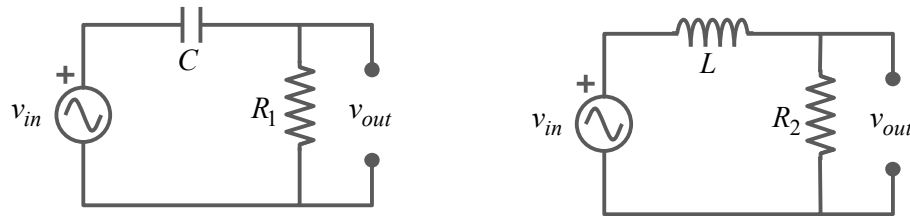


Figure Q8 : Series resistor-capacitor and resistor-inductor filtering circuits

- (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	$\int_{-\infty}^t x(\tau) d\tau$	$\frac{1}{j2\pi f} X(f) + \frac{1}{2} X(0) \delta(f)$
		$\frac{1}{j2\pi f} X(f) \text{ if } X(0) = 0$

**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	$t u(t)$	$1/s^2$
n <sup>th</sup> 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 <sup>st</sup> 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 <sup>nd</sup> 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 <sup>nd</sup> 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	
$\exp(\pm j\theta) = \cos(\theta) \pm j \sin(\theta)$	$\sin(\alpha \pm \beta) = \sin(\alpha) \cos(\beta) \pm \cos(\alpha) \sin(\beta)$
$\cos(\theta) = \frac{1}{2} [\exp(j\theta) + \exp(-j\theta)]$	$\cos(\alpha \pm \beta) = \cos(\alpha) \cos(\beta) \mp \sin(\alpha) \sin(\beta)$
$\sin(\theta) = \frac{1}{j2} [\exp(j\theta) - \exp(-j\theta)]$	$\tan(\alpha \pm \beta) = \frac{\tan(\alpha) \pm \tan(\beta)}{1 \mp \tan(\alpha) \tan(\beta)}$
$\sin^2(\theta) + \cos^2(\theta) = 1$	
$\sin(2\theta) = 2 \sin(\theta) \cos(\theta)$	$\sin(\alpha) \sin(\beta) = \frac{1}{2} [\cos(\alpha - \beta) - \cos(\alpha + \beta)]$
$\cos(2\theta) = \cos^2(\theta) - \sin^2(\theta)$	$\cos(\alpha) \cos(\beta) = \frac{1}{2} [\cos(\alpha - \beta) + \cos(\alpha + \beta)]$
$\sin^2(\theta) = \frac{1}{2} [1 - \cos(2\theta)]$	$\sin(\alpha) \cos(\beta) = \frac{1}{2} [\sin(\alpha - \beta) + \sin(\alpha + \beta)]$
$\cos^2(\theta) = \frac{1}{2} [1 + \cos(2\theta)]$	$C \cos(\theta) - S \sin(\theta) = \sqrt{C^2 + S^2} \cos \left[ \theta + \tan^{-1} \left( \frac{S}{C} \right) \right]$