THE UNIVERSITY OF AUCKLAND

FIRST SEMESTER, 2009 Campus: City

PHYSICS

Linear Systems

(Time allowed: THREE hours)

INSTRUCTIONS TO CANDIDATES

Answer any **FOUR** questions. All questions are of equal value.

> The otago paper PHSI461 and the Auckland University paper PHYSICS 701 were in 2009 (and still are) taught jointly using a video link.

> The exams sat were (and still are) the same in both institutions.

As far as I am aware in 2009 a Otago "branded" exam paper was not produced.

> -- Jevon Longdell, Physics Dept, March 2011.

A table of Fourier transforms and properties

DFT:
$$X[r] = \frac{1}{N} \sum_{k=0}^{N-1} x[k] \exp\left(-\frac{j2\pi rk}{N}\right)$$
 IDFT: $x[k] = \sum_{r=0}^{N-1} X[r] \exp\left(\frac{j2\pi rk}{N}\right)$

Hilbert transform:
$$\hat{f}(t) = f(t) * \frac{1}{\pi t}$$
, $\hat{F}(\nu) = -j \operatorname{sgn}(\nu) F(\nu)$

Convolution integral:
$$(f * h)(t) = \int_{-\infty}^{\infty} f(\tau) h(t - \tau) d\tau$$

(3 marks)

- 1. (a) Define what it means for a system to be *linear*, and to be *shift-invariant* (or time-invariant). (4 marks)
 - (b) Define the *impulse response* of a linear shift-invariant system, and show how the impulse response relates input and output of the system. (4 marks)
 - (c) Define the *transfer function* of a linear shift-invariant system, and give a formula relating the transfer function to the impulse response of the system. (4 marks)
 - (d) Define what is meant by the *stability* of a linear shift-invariant system and show that it is equivalent to the impulse response of the system being absolutely integrable. (6 marks)
 - (e) A system with an 'echo' may be represented by a linear-time invariant system with the impulse response

$$h(t) = \sum_{k=0}^{\infty} a^k \delta(t - kT)$$

where 0 < a < 1.

- (i) Is this system stable?
- (ii) Show that the transfer function $H(\nu)$ satisfies $H(v) = H(\nu + n/T)$ for all integers n. (4 marks)
- 2. (a) The function f(t) has a Fourier transform $F(\nu)$. Derive expressions, in terms of $F(\nu)$, for the Fourier transforms of

(i)
$$\frac{d}{dt}f(t)$$
, (ii) $f^*(t)$, and (iii) $\operatorname{Re}(f(t))$,

where * denotes a complex conjugate and Re(.) takes the real part of its argument. (6 marks)

- (b) Show that if $f(t) \leftrightarrow F(\nu)$ and $g(t) \leftrightarrow G(\nu)$ are Fourier transform pairs, then $(f * g)(t) \leftrightarrow F(\nu)G(\nu)$. (7 marks)
- (c) Find and sketch the Fourier transform of each of the functions:

$$w(t) = \begin{cases} 1 + \cos\left(\frac{\pi t}{M}\right), & |t| < M \\ 0 & \text{otherwise} \end{cases}$$
$$x(t) = \begin{cases} 1 - t^2, & |t| < 1 \\ 0 & \text{otherwise} \end{cases}$$
$$y(t) = \begin{cases} (1 - |t|)^2, & |t| < 1 \\ 0 & \text{otherwise} \end{cases}$$
$$z(t) = \begin{cases} (1 - |t|)^3, & |t| < 1 \\ 0 & \text{otherwise} \end{cases}$$

You may assume the transform pairs given on page 2.

(12 marks)

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3. A function f(t) is sampled with a uniform sampling interval of T seconds to give the sequence $\{f[k] : f[k] = f(kT)\}$, or equivalently the generalized function

$$f_s(t) = \sum_{k=-\infty}^{\infty} f[k]\delta(t-kT).$$

- (a) What is the relationship between the Fourier transforms of f(t) and $f_s(t)$? Illustrate your answer with diagrams as appropriate. (4 marks)
- (b) Using the result of part (a), or otherwise, derive Nyquist's sampling theorem, i.e. the conditions under which f(t) can be reconstructed from the sample values $\{f[k]\}$. For the case that those conditions are satisfied, write down (or derive) a relationship giving f(t) in terms of f[k]. (6 marks)
- (c) A simple, though approximate, method of reconstruction is to linearly interpolate between sample values, i.e.,

$$f_I(t) = f[k] + \alpha f[k+1]$$
 when $t = kT + \alpha T$ with $\alpha \in [0, 1]$.

Linear interpolation may be performed by a linear time-invariant system that produces output f_I for input f_s .

- (i) Give the impulse response of the interpolating system, and sketch the impulse response. (4 marks)
- (ii) Derive the transfer function of the interpolating system. (4 marks)
- (iii) Using your result from part (c)(ii), or otherwise, give the relationship between the Fourier transforms of the interpolated function f_I , and the original function f. Illustrate your answer with diagrams as appropriate. Under what condition is the linearly interpolated function f_I a good approximation to the original function f? (7 marks)

(6 marks)

- 4. (a) A modulating signal $m(t) = \cos(2\pi\nu_{\rm m}t)$ is transmitted using a carrier of frequency $\nu_{\rm c}$ $(\nu_{\rm c} \gg \nu_{\rm m})$. Write down expressions for the modulated signal f(t) and *sketch* its spectrum $F(\nu)$ for
 - (i) double sideband (DSB) modulation, and
 - (ii) upper sideband (USB) modulation.

Note: The Hilbert transform of $\cos(\omega t)$ is $\sin(\omega t)$ ($\omega > 0$).

(b) Two band-limited real signals $m_1(t)$ and $m_2(t)$ are applied to the quadrature modulator shown below, where the carrier frequency is ν_c . The spectra $M_1(\nu)$ and $M_2(\nu)$ vanish for $|\nu| \ge \nu_m$.



- (i) Write down expressions for the transmitted signal f(t) and its spectrum $F(\nu)$. (3 marks)
- (ii) At the receiver, the signal f(t) is multiplied by a reconstructed carrier $\cos(2\pi\nu_c t + \phi)$, and the output is low-pass filtered, so that only frequencies satisfying $|\nu| < \nu_f$ are passed through. Calculate the output signal g(t) for an arbitrary value of ϕ , and show that it is possible to recover $m_1(t)$ and $m_2(t)$ by making appropriate choices for ϕ . What relationships must hold between ν_m , ν_c , and ν_f for this recovery to be possible? (6 marks)

Note: $\cos(\theta)\cos(\phi) = \frac{1}{2}[\cos(\theta + \phi) + \cos(\theta - \phi)]$ and $\sin(\theta)\cos(\phi) = \frac{1}{2}[\sin(\theta + \phi) + \sin(\theta - \phi)]$

- (c) (i) A finite energy signal f(t) has Fourier transform $F(\nu)$. By considering the passage of this signal through an ideal band-pass filter, followed by an energy measurement, show that $|F(\nu)|^2$ gives the energy spectral density. (5 marks)
 - (ii) Define the *energy autocorrelation function* of the signal f(t), and show that the energy spectral density is the Fourier transform of this function. (5 marks)

- 5. (a) (i) Write down the definition of the two-sided Laplace transform $F_L(s)$ of the function f(t) and describe what is meant by the region of absolute convergence of the Laplace transform. Write down the corresponding inverse Laplace transform relationship as a line integral in the complex plane. (5 marks)
 - (ii) Using the complex inversion formula, find the inverse Laplace transform of

$$F_L(s) = \frac{1}{(s-1)(s+2)},$$

where the region of convergence is $-2 < \operatorname{Re}(s) < 1.$ (8 marks)

(b) The coordinate x(t) of a *driven damped harmonic oscillator* is described by the differential equation

$$\frac{\mathrm{d}^2 x(t)}{\mathrm{d}t^2} + \gamma \frac{\mathrm{d}x(t)}{\mathrm{d}t} + \omega^2 x(t) = f(t) \,,$$

where γ is the damping rate, $\omega/(2\pi)$ is the natural frequency of the oscillator, and f(t) is the (time-dependent) driving force. Given the initial conditions

$$x(t=0) = 0$$
, $\frac{\mathrm{d}x}{\mathrm{d}t}\Big|_{t=0} = 0$,

use Laplace transforms to show that

$$x(t) = \frac{1}{p_{+} - p_{-}} \int_{0}^{t} \left[e^{p_{+}(t-\tau)} - e^{p_{-}(t-\tau)} \right] f(\tau) \, \mathrm{d}\tau \,,$$

where

$$p_{\pm} = -\frac{\gamma}{2} \pm \frac{1}{2}\sqrt{\gamma^2 - 4\omega^2} \,.$$

(12 marks)

- 6. (a) A plane wave of light of wavelength λ and unit amplitude is normally incident on a screen which is opaque except for a circular aperture of radius a.
 - (i) Use the paraxial diffraction integral to write down an expression for the amplitude of the light at a point $(x_{\rm P}, y_{\rm P})$ on a plane located at distance d on the other side of the screen. (Note: You do not have to evaluate the integral.) (2 marks)
 - (ii) Specialise now to the point P lying on the z-axis (i.e., $x_P = y_P = 0$). Evaluate the integral (using polar coordinates) and calculate the intensity. Show that the intensity at P is a minimum when $a^2 = 2n\lambda d$, where n is an integer. (9 marks)

Note: The paraxial diffraction integral is

$$f_z(x,y) = \frac{1}{j\lambda z} \int_{-\infty}^{\infty} \int_{-\infty}^{\infty} f_0(x_0, y_0) \exp\left\{\frac{jk}{2z} \left[(x - x_0)^2 + (y - y_0)^2\right]\right\} dx_0 dy_0,$$

where $k = 2\pi/\lambda$.

- (b) Describe the *Fraunhofer approximation* to diffraction from an aperture. What is its region of validity for the circular aperture described above? (5 marks)
- (c) Consider a Gaussian beam, which at z = 0 is described by $f_0(x, y) = A_0 \exp[-B_0(x^2 + y^2)]$, where A_0 and B_0 are *real*. In the paraxial approximation, show that the beam remains Gaussian as it propagates, but is described by the form $f_z(x, y) = A(z) \exp[-B(z)(x^2 + y^2)]$, where A(z) and B(z) are complex for $z \neq 0$. Give the expressions for A(z) and B(z) and show that the *half-width* of the beam, w(z), is given by

$$w(z) = \sqrt{\frac{1 + 4B_0^2 z^2/k^2}{B_0}}$$

(9 marks)

Note: The Fourier transform of the paraxial diffraction integral is

$$F_z(u,v) = F_0(u,v) \exp\left[-j\frac{2\pi^2}{k}(u^2+v^2)z\right],$$

and $\exp(-ax^2) \leftrightarrow \sqrt{\pi/a} \exp(-\pi^2 u^2/a)$.