## **ENGINEERING PHYSICS 3W4**

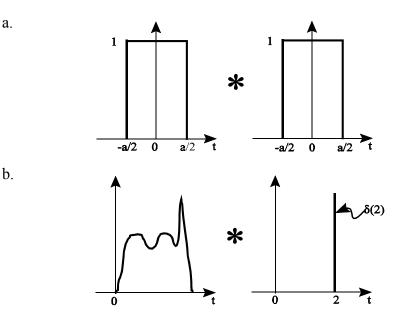
DAY CLASS	Dr. Wm. Garland
DURATION: 2 hours	Page 1 of 3
McMASTER UNIVERSITY FINAL EXAMINATION	April 2000
Special Instructions:	

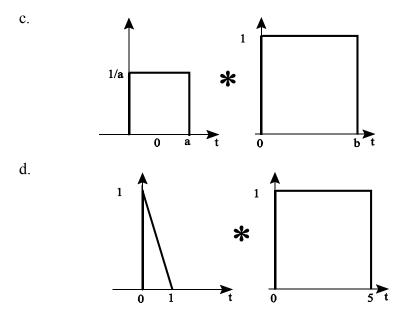
## 1. Closed Book. All calculators and up to 6 single sided (or 3 double sided) 8 <sup>1</sup>/<sub>2</sub>" by 11" crib sheets are permitted.

- 2. Do all questions.
- 3. The value of each question is as indicated.
- 4. Point form is sufficient for discussion type questions. **TOTAL Value**: 100 marks

## THIS EXAMINATION PAPER INCLUDES 3 PAGES AND 7 QUESTIONS. YOU ARE RESPONSIBLE FOR ENSURING THAT YOUR COPY OF THE PAPER IS COMPLETE. BRING ANY DISCREPANCY TO THE ATTENTION OF YOUR INVIGILATOR.

- 1. [15 marks] Define:
  - a. Linear system
  - b. Causal system
  - c. Time invariant system
  - d. Convolution integral
  - e. Autocorrelation
- 2. [20 marks] Graphically compute the convolution of the following function pairs:





3. [15 marks] Evaluate the following expressions:

a. Evaluate 
$$\int_{-\infty}^{+\infty} \delta(-2t-1) \operatorname{sinc}(\pi vt) dt$$

b. Evaluate 
$$\frac{d}{dt}\Pi_a(t)$$

c. Show that 
$$f(t) \neq (\frac{1}{2\pi}) \frac{dF(v)}{dv}$$
. [Don't panic; it's easy!]

4. [10 marks] Give brief answers to the following:

- a. In signal analysis, what is the advantage of reflecting a time signal about the origin?
- b. Show how the convolution integral results from making a measurement. Assume a linear, time invariant, causal system.
- c. Why are Fourier Transforms so useful when dealing with convolutions?
- d. Laplace and Fourier Transforms are similar. Why are Laplace transforms used more for solving differential equations and Fourier Transforms used more for signal analysis?
- 5. [10 marks] In the following, a proof is not required.
  - a. What is the FT of  $[a(t)*b(t) + c(t).d(t)]^2$
  - b. What is the FT of  $\Pi_2(t)$ . Graph the result.
  - c. What is the FT of  $exp(-\pi t^2)$ ?

- 6. [20 marks] An input step voltage,  $V_0$  u(t), is passed through an ideal low pass filter.  $V_0$  is the magnitude of the voltage and u(t) denotes the unit step.
  - a. Sketch the input signal in time space.
  - b. What does the ideal low-pass filter look like in frequency space?
  - c. What does the ideal low-pass filter look like in time space?
  - d. By working only in time space, determine the output voltage.
  - e. What is the Fourier Transform of the input  $V_0$  u(t)? We denote this as  $V_0 \varphi(v)$
  - f. What is the output voltage in frequency space, Y(v)?
  - g. By performing the Inverse Fourier Transform of Y(v), determine the output voltage in time space, y(t).
  - h. Referring to both the time space and the frequency space, describe briefly what happens to the signal as it passes through the filter.
- 7. [10 marks] We have shown that a sampled signal in frequency space is

$$\hat{\mathbf{S}}(\mathbf{v}) = \sum_{n=-\infty}^{+\infty} \mathbf{s}(nT) e^{-2\pi i \mathbf{v} nT}.$$

- a. Show that is a periodic function.
- b. Discuss the ramifications for setting the sampling rate.
- c. Find the critical sampling rate.

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