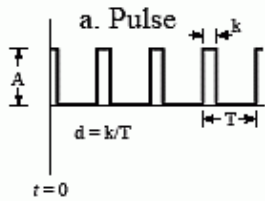


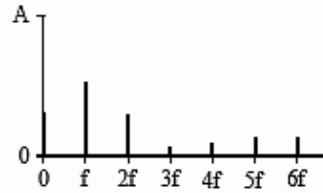
**EE354 Study Guide for Exam #1**

**Provided Reference Sheets**

**Time Domain**



**Frequency Domain**

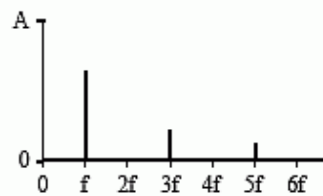
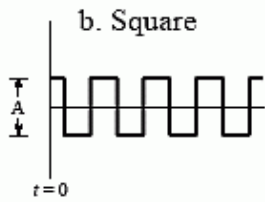


$$a_0 = A d$$

$$a_n = \frac{2A}{n \pi} \sin(n \pi d)$$

$$b_n = 0$$

( $d = 0.27$  in this example)

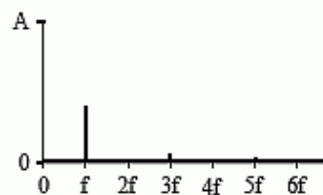
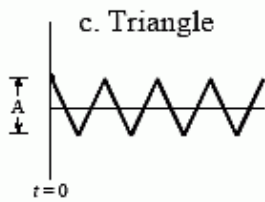


$$a_0 = 0$$

$$a_n = \frac{2A}{n \pi} \sin\left(\frac{n \pi}{2}\right)$$

$$b_n = 0$$

(all even harmonics are zero)

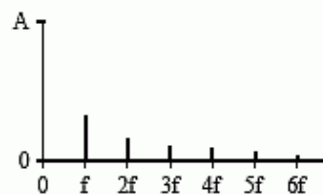
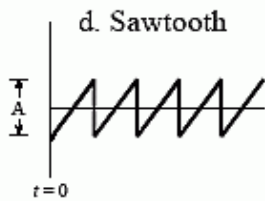


$$a_0 = 0$$

$$a_n = \frac{4A}{(n \pi)^2}$$

$$b_n = 0$$

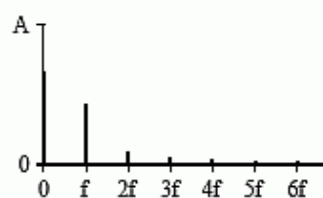
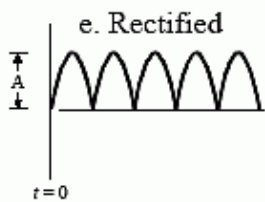
(all even harmonics are zero)



$$a_0 = 0$$

$$a_n = 0$$

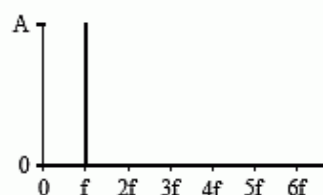
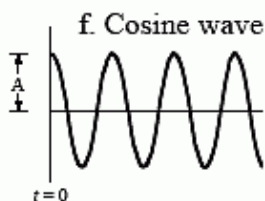
$$b_n = \frac{A}{n \pi}$$



$$a_0 = 2A/\pi$$

$$a_n = \frac{-4A}{\pi(4n^2 - 1)}$$

$$b_n = 0$$



$$a_1 = A$$

(all other coefficients are zero)

FIGURE 13-10 Examples of the Fourier series. Six common time domain waveforms are shown, along with the equations to calculate their "a" and "b" coefficients.

**Workbook Tables (B.3 and B.4)**

$\Pi\left(\frac{t}{\tau}\right) \Leftrightarrow \tau \operatorname{sinc}(f\tau)$	$ax(t) + by(t) \Leftrightarrow aX(f) + bY(f)$	Superposition (or Linearity)
$e^{j\omega_0 t} \Leftrightarrow \delta(f - f_0)$	$X(t) \Leftrightarrow x(-f)$	Duality
$\delta(t - t_0) \Leftrightarrow e^{-j2\pi f t_0}$	$\sum_{n=-\infty}^{\infty} c_n e^{jn\omega_0 t} \Leftrightarrow \sum_{n=-\infty}^{\infty} c_n \delta(f - n f_0)$	F.T. of periodic function
$\sum_{k=-\infty}^{\infty} \delta(t - kT_0) \Leftrightarrow f_0 \sum_{k=-\infty}^{\infty} \delta(f - k f_0)$	$\frac{d^n}{dt^n} x(t) \Leftrightarrow (j2\pi f)^n X(f)$	Differentiation in time
$\delta(t) \Leftrightarrow 1$	$x(t - t_0) \Leftrightarrow e^{-j2\pi f t_0} X(f)$	Time shifting
$K \Leftrightarrow K \delta(f)$	$x(at) \Leftrightarrow \frac{1}{ a } X\left(\frac{f}{a}\right)$	Time scaling
$\operatorname{sinc}(2Wt) \Leftrightarrow \frac{1}{2W} \Pi\left(\frac{f}{2W}\right)$	$\int_{-\infty}^t x(\lambda) d\lambda \Leftrightarrow \frac{1}{j2\pi f} X(f) + \frac{X(0)}{2} \delta(f)$	Integration
$\cos(\omega_0 t) \Leftrightarrow \frac{1}{2} \delta(f - f_0) + \frac{1}{2} \delta(f + f_0)$	$x(t) * y(t) \Leftrightarrow X(f)Y(f)$	Convolution
$\sin(\omega_0 t) \Leftrightarrow \frac{1}{2j} \delta(f - f_0) - \frac{1}{2j} \delta(f + f_0)$	$x(t)y(t) \Leftrightarrow X(f) * Y(f)$	Multiplication
$u(t) \Leftrightarrow \frac{1}{j2\pi f} + \frac{1}{2} \delta(f)$	$e^{j\omega_0 t} x(t) \Leftrightarrow X(f - f_0)$	Frequency shifting
$\Lambda\left(\frac{t}{\tau}\right) \Leftrightarrow \tau \operatorname{sinc}^2(f\tau)$	$t^n x(t) \Leftrightarrow (-j2\pi)^{-n} \frac{d^n}{df^n} X(f)$	Differentiation in frequency
$\sum_{k=-\infty}^{\infty} \Pi\left(\frac{t - kT_0}{\tau}\right) \Leftrightarrow \tau f_0 \sum_{k=-\infty}^{\infty} \operatorname{sinc}(kf_0\tau) \delta(f - kf_0)$	$x(t) \sum_{k=-\infty}^{\infty} \delta(t - kT_s) \Leftrightarrow f_s \sum_{k=-\infty}^{\infty} X(f - kf_s)$	Impulse sampling
$\operatorname{sinc}^2(2Wt) \Leftrightarrow \frac{1}{2W} \Lambda\left(\frac{f}{2W}\right)$	$x(t) \cos(\omega_c t) \Leftrightarrow \frac{1}{2} X(f + f_c) + \frac{1}{2} X(f - f_c)$	Modulation
$e^{-at} u(t) \Leftrightarrow \frac{1}{a + j2\pi f}$		
$te^{-at} u(t) \Leftrightarrow \frac{1}{(a + j2\pi f)^2}$		

## **Fourier Series**

- Given an *arbitrary* time-domain signal, compute the trigonometric and exponential Fourier Series Coefficients, write the Fourier Series approximation of the signal, and draw the Fourier Spectrum.
- Given an *arbitrary* set of Fourier Coefficients, draw the Fourier Spectrum and write the time domain trigonometric or exponential Fourier Series.
- Given a Fourier Series or Fourier Spectrum of an arbitrary signal, compute the Fourier Series or Fourier Spectrum after passing the signal through a high-pass, low-pass, or bandpass filter.

## **Fourier Transforms**

- Know (memorize) the following Fourier Transform Pairs:
  - a. Rect / Sinc
  - b. Delta / Constant
  - c. Cosine / Even Delta
  - d. Sine / Odd Delta
  - e. Triangle / Sinc<sup>2</sup>
  - f. Impulse Train / Impulse Train
- Given an *arbitrary* signal in the time-domain, use the provided tables to compute the Fourier Transform and draw the signal spectrum.
- Given an *arbitrary* signal spectrum, use the provided tables to compute the Inverse Fourier Transform and write the signal in the time-domain.

## **Convolution**

- Given two *arbitrary* (but simple) signals,  $a(t)$  and  $b(t)$ , compute the result of  $a(t)*b(t)$ .
- Apply the properties of convolution in the frequency domain to simplify and solve convolution of  $a(t)*b(t)$ .
- Understand and be able to apply the following convolution properties:
  - a. Identity
  - b. Commutative
  - c. Associative
  - d. Distributive
- Apply the properties of convolution with special functions (e.g., delta-functions or constants) to simplify and solve convolution of an arbitrary signal with the special function.

## **Energy/Power and Filtering**

- Convert between gain as a linear ratio and gain expressed as a dB quantity.
- Convert between powers expressed as Watts and powers expressed as dBm.
- Use Parseval's Theorem to calculate the total energy/power in a signal given either the time-domain or frequency-domain representation of the signal.
- Given an *arbitrary* signal, apply the frequency domain version of convolution to determine the output (time-domain or frequency domain) of a high-pass, low-pass, or bandpass filter.

## Sampling Theorem

- Explain the difference between natural, flat-top, and impulse sampling a signal.
- For an arbitrary signal (expressed in either the time-domain or as a frequency spectrum), calculate the Nyquist sampling frequency.
- Define aliasing and explain how aliasing is related to the Nyquist sampling frequency. Calculate the output spectrum when a signal is sampled at a rate that would cause aliasing.
- For an arbitrary signal which is impulse sampled at a sampling frequency of  $f_s$ , apply the sampling theorem and calculate the frequencies present in the sampled signal's spectrum.
- Draw the frequency spectrum for an arbitrary signal that is impulse sampled at a sampling frequency of  $f_s$ .
- Apply a filter to the sampled signal and determine the time domain and frequency domain output of the filter.

## A/D Conversion and PCM

- Define Quantization and explain how quantization introduces error into a sampled signal.
- Given an ADC (analog-to-digital converter), determine the number of bits of quantization required to obtain a specified SQNR (i.e., Dynamic Range); given the number of bits for an ADC, determine the resulting SQNR.
- Given an ADC with a specified full-scale input range ( $V_{fs}$ ) and number of bits of quantization, determine the quantization error (resolution) of the ADC.
- Given an arbitrary signal, design a PCM modulation system by calculating the required sampling frequency and number of bits of quantization necessary to achieve a specified SQNR.
- Given a PCM system, calculate the bit rate necessary to transmit the PCM encoded signal.

## Amplitude Modulation

- Explain in words the difference between DSB-SC, DSB-TC, and SSB Amplitude Modulation.
- Given an arbitrary modulating signal, express mathematically a DSB-SC and DSB-TC signal in both the time-domain and frequency-domain.
- Given an arbitrary modulating signal, illustrate the resulting Amplitude Modulated signal in both the time-domain and frequency-domain.
- Given an arbitrary modulating signal, calculate the bandwidth of the resulting DSB-TC, DSB-SC, or SSB AM signal.
- Given a sinusoidally modulated DSB-TC AM signal, calculate the modulation index, carrier power, sideband power, total power, and power efficiency.
- Explain in words how to demodulate a DSB-TC AM signal with a modulation index less than 1.0.

## Frequency Modulation

- Explain in words the difference between Amplitude Modulation and Angle Modulation.
- Explain in words the difference between Frequency Modulation and Phase Modulation.
- For an arbitrary signal, calculate the instantaneous frequency of the signal using the definition of instantaneous frequency.
- Explain in words the difference between narrowband FM and wideband FM.
- Given an arbitrary modulating signal, express mathematically a PM or FM signal in both the time-domain and frequency-domain.
- Given an arbitrary modulating signal, illustrate the resulting FM or PM signal in both the time-domain and frequency-domain.
- Given a sinusoidally modulated FM or PM signal, calculate the modulation index, total power in the signal, power in a given sideband, and power efficiency.
- Given a sinusoidally modulated FM or PM signal, calculate both the Absolute Bandwidth and Carson's Rule Bandwidth. Explain in words the difference between the two bandwidths.