For exam 2, you need to know (1) impulse response, (2) linearity and time invariance, (3) frequency response, (4) discrete-time processing of continuous-time periodic signals, (5) DTFT. In other words, you should be able to do the following problems.

2.1 Impulse Response

Problem:

A particular system computes the average of 5 samples, minus the average of the previous five:

$$y[n] = \frac{1}{5} \sum_{m=0}^{4} x[n-m] - \frac{1}{5} \sum_{m=5}^{9} x[n-m]$$

Find the impulse response h[n].

Solution:

Feed $\delta[n]$ into the system, and the output is:

$$h[n] = \begin{cases} \frac{1}{5} & 0 \le n \le 4\\ -\frac{1}{5} & 5 \le n \le 9\\ 0 & \text{otherwise} \end{cases}$$

2.2 Linearity and Time Invariance

Problem:

A particular system changes the sign of every second input sample, thus

$$y[n] = (-1)^n x[n]$$

Is this system linear? It is time-invariant? Prove your answers. **Problem:**

$$\begin{array}{rccc} x_1[n] & \to & (-1)^n x_1[n] \\ x_2[n] & \to & (-1)^n x_2[n] \\ ax_1[n] + bx_2[n] & \to & (-1)^n (ax_1[n] + bx_2[n]) \\ & = & ay_1[n] + by_2[n] \end{array}$$

So the system is linear.

$$\begin{array}{rcl} x_1[n] & \to & (-1)^n x_1[n] \\ y_1[n-m] & = & (-1)^{n-m} x_1[n-m] \\ x_1[n-m] & \to & (-1)^n x_1[n-m] \neq y_1[n-m] \end{array}$$

So the system is not time-invariant.

2.3 DT Processing of CT Signals

Problem:

A continuous-time periodic signal x(t) with a period of $T_0 = 0.03$ seconds is passed through an ideal anti-aliasing filter $H_a(\Omega)$, then sampled at $F_s = 1000$ Hz to produce the discrete-time signal x[n]. The discrete-time signal x[n] is then processed by the following system:

$$y[n] = \frac{1}{10} \sum_{m=0}^{9} x[n-m]$$

The signal y[n] is then passed through an ideal D/A, at the same sampling frequency $F_s = 1000$ Hz, to produce the signal y(t). Let Y_k be the Fourier series coefficients of y(t), and X_k those of x(t).

- 1. Assume that $X_k \neq 0$ for all k; identify all k for which $Y_k = 0$.
- 2. Specify $|Y_k|$ in terms of $|X_k|$ for all non-zero Y_k .

Solution:

The harmonic frequencies of both x(t) and y(t) are $k\Omega_0 = 2\pi k/T_0 = 200\pi/3$ radians/second. The antialiasing filter eliminates all components $|\Omega| \ge \pi F_s = 1000\pi$, thus, all harmonics with $|k| \ge 15$. The other harmonics are mapped to $k\omega_0 = k\Omega_0/F_s = 2\pi k/30$. The discrete-time system has the frequency response

$$H(\omega) = e^{-j\omega(10-1)/2} \frac{\sin(\omega 10/2)}{10\sin(\omega/2)}$$

which has zeros at $\omega_{\ell} = 2\pi \ell/10$ for $\ell \neq 0$, thus every third harmonic of x[n] is zeroed out. Thus

$$|Y_k| = \begin{cases} 0 & k = 3\ell, \ \ell \neq 0\\ 0 & |k| \ge 15\\ \left|\frac{\sqrt{3}}{20\sin(k\pi/30)}X_k\right| & \text{otherwise} \end{cases}$$

2.4 DTFT

Problem:

A particular bandpass filter has the frequency response

$$F(\omega) = \begin{cases} 1 & \frac{\pi}{10} \le |\omega| \le \frac{\pi}{4} \\ 0 & \text{otherwise} \end{cases}$$

Find the impulse response f[n].

Solution:

If your formula sheet has the form of an ideal lowpass filter on it, then the easiest way to solve this problem is to notice that

$$F(\omega) = G(\omega) - H(\omega)$$

where $G(\omega)$ is an ideal LPF with a cutoff of $\pi/4$, and $H(\omega)$ is an ideal LPF with a cutoff of $\pi/10$, thus

$$f[n] = \frac{\sin(\pi n/4) - \sin(\pi n/10)}{\pi n}$$

Alternatively, you could use the inverse DTFT formula directly:

$$f[n] = \frac{1}{2\pi} \int_{-\pi/4}^{-\pi/10} e^{j\omega n} d\omega + \frac{1}{2\pi} \int_{\pi/10}^{\pi/4} e^{j\omega n} d\omega$$
$$= \frac{1}{2\pi jn} \left[e^{j\pi n/4} - e^{j\pi n/10} + e^{-j\pi n/10} - e^{j\pi n/4} \right] = \frac{\sin(\pi n/4) - \sin(\pi n/10)}{\pi n}$$