

EEE2035F EXAM SIGNALS AND SYSTEMS I

HINTS: June 2015

1. Here (a) is a shift to the right by 2, (b) is time reversed with a subsequent shift to take the origin to $t = 2$, (c) involves compression of the signal by a factor of two around the origin, coupled by an overall scaling of the range by -2 (i.e. upside down and bigger), (d) involves compression of the signal by a factor of two around the origin followed by a shift to take the origin to $t = 1$, (e) is the indefinite integral (0 before $t = -1$, 2 at $t = 1$, 1.5 for $t > 2$, with linear pieces in between), and (f) is zero everywhere except for delta functions at $t = -1$, $t = 1$ and $t = 2$ with sizes 1, -1.5 , and 1.5 respectively.
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2. (a) Impulse response $h(t)$ is output when input is Dirac delta, or $i(t) = \delta(t)$, so $h(t) = u(t)$.
(b) To calculate output at $t = t_0$ require knowledge of input for $-\infty < t \leq t_0$. These values all in past, so causal, and memory is required. Could also note that impulse response $h(t)$ is zero for $t < 0$ so causal.
(c) Supposing input $i(t) = i_0(t)$ gives output $q_0(t) = \int_{-\infty}^t i_0(\phi) d\phi$. Output for $i(t) = i_0(t - c)$ is $q(t) = \int_{-\infty}^t i_0(\phi - c) d\phi = \int_{-\infty}^{t-c} i_0(\psi) d\psi = q_0(t - c)$ so time invariant.
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3. With $z(t) = h(t) * u(t)$ the derivative property states that $\dot{z}(t) = h(t) * u(t) = h(t) * \delta(t) = h(t)$, so can find $z(t)$ by indefinite integration. Then use $y(t) = h(t) * u(t - 1) = z(t - 1)$ and plot as required.
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4. (a) Exponential decaying to the right, truncated to start at $t = -1$.
(b) Impulse response is derivative of step response. Zero for $t < -1$ and turns out

to be negative for $t > -1$. Remember delta function at $t = -1$ to take care of discontinuity.

(c) Impulse response not zero for all $t < 0$ so not causal.

(d) The given $g(t)$ is the output when the input is $u(t)$. The input $3u(t + 4)$ is just a scaled and shifted version of the previous input, so the output in this case will be $3g(t + 4)$.

5. (a) Just substitute and get values of c_k for $k \geq 0$. The values will in this instance be real, but nonetheless just plot them in the Argand diagram, read off the magnitude and phase, and plot accordingly. Since $x(t)$ is real the coefficients will satisfy $c_k = c_{-k}^*$ (i.e. magnitudes equal but phases will have opposite sign).

(b) Third harmonic is $c_{-3}e^{-j3\pi t} + c_3e^{j3\pi t}$. By Parseval's theorem the total power will be $|c_{-3}|^2 + |c_3|^2 = 2|c_3|^2$ which can easily be calculated.

(c) Here $y(t) = 2x(t - 0.5) = 2 \sum_{-\infty}^{\infty} c_k e^{jk\pi(t-0.5)} = \sum_{-\infty}^{\infty} 2c_k e^{-jk\pi 0.5} e^{jk\pi t}$, which is the form of a Fourier series with coefficients $d_k = 2c_k e^{-jk\pi/2}$ and the values can be found.

6. (a) Take Fourier transforms of both sides, use the derivative property, and manipulate to find $Y(\omega)/X(\omega) = H(\omega)$.

(b) Use $|H(\omega)|^2 = H(\omega)H^*(\omega)$ to find $|H(\omega)| = \omega RC / \sqrt{1 + (\omega RC)^2}$. Observe that $|H(0)| = 0$ and $|H(\omega)| \rightarrow 1$ as $\omega \rightarrow \pm\infty$ and plot accordingly. This is a first-order highpass filter.

(c) Frequency response can be written as $H(\omega) = a + b / (\frac{1}{RC} + j\omega)$. Solve for unknown coefficients and invert for impulse response

$$h(t) = \delta(t) - \frac{1}{RC} e^{-\frac{1}{RC}t} u(t).$$

(d) Input is $x(t) = 1/2 e^{j5\pi t} + 1/2 e^{-j5\pi t}$ so using fundamental properties of the frequency response the output will be $y(t) = \frac{1}{2} H(5\pi) e^{j5\pi t} + \frac{1}{2} H(-5\pi) e^{-j5\pi t}$. Find $H(5\pi)$ and $H(-5\pi)$ (they will be conjugates) and the result can be written as a real cosine again if required.

7. (a) Use frequency translation on pair $e^{-3t} u(t) \xleftrightarrow{\mathcal{F}} \frac{1}{3+j\omega}$ and then time shifting to make the right-hand side look like $X(\omega)$. The left-hand side is then the required

$$x(t) = 6e^{-3(t+1)}u(t+1)e^{j4(t+1)}.$$

- (b) Time scaling to the given transform provides $e^{-|at|} \xrightarrow{\mathcal{F}} \frac{2a}{a^2 + \omega^2}$. Then apply duality followed by frequency shift to get $X(\omega) = \frac{\pi}{a}e^{-|a(\omega-W)|}$.

8. (a) The required spectrum will contain two half-amplitude replicas of $G(\omega)$ centered at $\omega = \pm 20$ rad/s.
- (b) The given system is an ideal lowpass filter that keeps content between $-20 \leq \omega \leq 20$, so the "outside" halves of each replica in $G(\omega)$ will be eliminated. Easiest to see graphically but you should get $Q(\omega) = 1/2p_{10}(\omega + 15) + 1/2p_{10}(\omega - 15)$.
- (c) Total energy is $E = 1/(2\pi) \int_{-\infty}^{\infty} |Q(\omega)|^2 d\omega = 2(10/4)/(2\pi)$ (Joules).
- (d) Since $\omega_{max} = 10$ and must sample with a frequency $\omega = 2\pi/T > 2\omega_{max}$, require $T < \pi/10$ seconds.
- (e) Pulse train $g(t)$ has transform $P(\omega) = 2\pi/T_s \sum_{k=-\infty}^{\infty} \delta(\omega - k\omega_s)$ with $\omega_s = 2\pi/T_s = 30$ rad/s. Multiplication in time is convolution in frequency, so $G_s(\omega) = 1/T_s \sum_{k=-\infty}^{\infty} G(\omega - k\omega_s)$ is the required result and contains scaled replicas of $G(\omega)$ at spacings ω_s .