1. Determine the fundamental period of x[n].

$$x[n] = \cos(\frac{n\pi}{10}) + \sin(\frac{n\pi}{15})$$

2. Determine whether or not the signal x[n] is periodic.

$$x[n] = \sin(\sqrt{2} + 0.2n)$$

3. Given that real valued signal $x_1[n]$ is even by definition $x_1[n] = x_1[-n]$, and real valued signal $x_2[n]$ is odd by definition $x_2[n] = -x_2[-n]$, determine symmetricality (even/odd) of y[n].

$$y[n] = x_1[n] \cdot x_2[n]$$

4. Given that the power of real valued signal x[n] is defined as $P=\sum_{n=-\infty}^{\infty}x^2[n]$, compute the power in y[n].

$$y[n] = 2^n \cdot u[-n]$$

5. Given that x[n] is the system input and y[n] is the system output, determine whether or not the following systems is time (shift)-invariant.

$$y[n] = x[n] \cdot u[n]$$

6. Given that x[n] is the system input and y[n] is the system output, determine whether or not the following systems is linear.

$$y[n] = \operatorname{Im} \big(x[n] \big)$$

 Given that x[n] is the system input and y[n] is the system output, determine whether or not the following systems is casual.

$$y[n]=x[|n|]$$

 Given that x[n] is the system input and y[n] is the system output, determine unit sample response h[n] of the system.

$$y[n] = 0.5y[n-1] + 4x[n-2]$$

9. The responses of a linear time (shift)-invariant system to specified inputs are defined as follows:

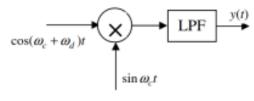
input	reponse		
name	symbol	reponse	
unit sample	$\delta[n]$	h[n]	
unit step	u[n]	s[n]	

calculate h[n] for the system, given that s[n] = u[n] - u[n-5].

10. Find the Fourier transform of x(t).

$$x(t) = \begin{cases} \frac{1}{2}, & -T < t < T \\ 0, & other \end{cases}$$

 Given that the cut-off frequency of the low pass filter (LPF) is w_c evaluate the output y(t).
 Note: LPF allows frequency values between -w_c < w < w_c.



$$x[n]=2\delta[n+2]-\delta[n+1]+3\delta[n]-\delta[n-1]+2\delta[n-2]$$

12. Given that $x[n] \stackrel{DTFT}{\Longleftrightarrow} X(e^{jw})$ is a DTFT pair, evaluate $X(e^{jw})|_{w=\pi}$ without explicitly finding $X(e^{jw})$.

$$x[n]=2\delta[n+2]-\delta[n+1]+3\delta[n]-\delta[n-1]+2\delta[n-2]$$

13. Given that $x[n] \stackrel{DTFT}{\Longleftrightarrow} X(e^{jw})$ and $y[n] \stackrel{DTFT}{\Longleftrightarrow} Y(e^{jw})$ are DTFT pairs, prove the convolution theorem.

$$x[n]*y[n] \xleftarrow{DTFT} X(e^{jw})Y(e^{jw})$$

14. Find the inverse DTFT of $X(e^{jw})$.

$$X(e^{jw}) = \cos^2(w)$$

15. Given the 6-point sequence x[n] = [4, -1, 4, -1, 4, -1], determine its 6-point DFT sequence X[k].

16. If the 4-point DFT an unknown length-4 sequence v[n] is V[k]=[1,4+j,-1,4-j], determine v[n].

17. Find z-transforms of x[n].

$$x[n] = 6\delta[n] - 7\delta[n-3] - 2\delta[n] - 9\delta[n-5]$$

18. If the region of convergence (ROC) for any x[n]

x[n] X(z) z-transform pair includes the unit cirle in the complex plane then, x[n] X(e^{jw}) DTFT pair can also be calculated (converges). Given that the following X(z) includes the unit circle in its region of convergence, evaluate DTFT of x[n] at w = π.

$$X(z) = \frac{z + 2z^{-2} + z^{-3}}{1 - 3z^{-4} + z^{-5}}$$

19. Evaluate h[n]*x[n] using the convolution property of z-transform.

$$h[n] = (0.5)^n u[n]$$

$$x[n] = 3^n u[-n]$$

20. Given that $x[n] \stackrel{ZT}{\Longleftrightarrow} X(z)$ is a z-transform pair find x[n].

$$X(z) = 2 + 5(z^2 + z^{-2})$$

21. Given that $x[n] \stackrel{ZT}{\Longleftrightarrow} X(z)$ is a z-transform pair find x[n] for |z| > 2.

$$X(z) = \frac{1}{1+3z^{-1}+2z^{-2}}$$

22. A continuous-time sinusoid a₁(t) = cos(w₁t + 0.1π) is sampled at f_{s1} = 40Hz to give a₁[n], and a second continuous-time sinusoid a₂(t) = cos(w₂t + 0.1π) is sampled at f_{s2} = 50 Hz to give a₂[n]. If w₂ = 30π rad/s, determine w₁ so that a₁[n] = a₂[n]. Assume there is no aliasing when sampling a₁(t).

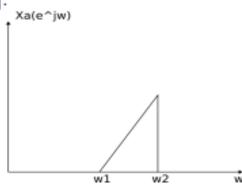
23. A complex bandpass filter is constructed by frequency shifting a running sum filter. evaluate and plot the magnitude frequency response of the complex bandpass filter |H_B(e^{jw})|.

$$h[n] = \sum_{k=0}^{4} \delta[n-k]$$

$$h_B[n] = h[n]e^{jw_0n}$$

$$h_B[n] \stackrel{DTFT}{\Longleftrightarrow} H_B(e^{jw})$$

24. A complex bandpass analog signal $x_a(t)$ has Fourier transform that is non-zero over the range of $[w_1, w_2]$. The signal is sampled to produce the sequence $x[n] = x_a(nT_s)$. What is the smallest sampling frequency that can be used so that $x_a(t)$ may be recovered from its samples x[n].



25. Plot STFT (Short Time Fourier Transform) representation of a 1D chirp signal with different window sizes. Compare the results with FT of the same signal. Comment on what might be an optimal window size.