

Applied Signal Processing – 17/4/2007 9.15-12.15h VERSION B

1. Find the DTFT of the two-sided sequence $x[n] = \left(\frac{1}{3}\right)^{|n|}$. (1)

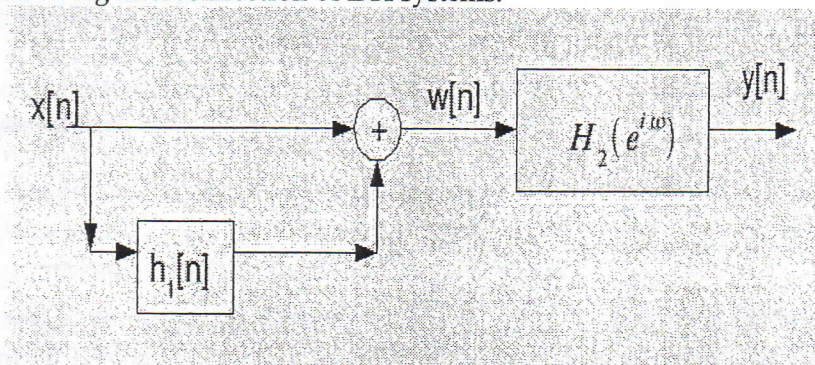
2. Find the z-transform of the following sequences. Wherever convenient, use the properties of the z-transform to make the solution easier:

(a) $x[n] = \left(\frac{1}{3}\right)^n \mu[-n]$ (0.5)

(b) $x[n] = \left(\frac{1}{2}\right)^n \mu[-n+2] + 3^n \mu[n-1]$ (0.5)

(c) $x[n] = 2n \left(\frac{1}{2}\right)^n \mu[n+1]$ (0.5)

3. Consider the following interconnection of LTI systems:



where $h_1[n] = \delta[n-1]$ and

$$H_2(e^{j\omega}) = \begin{cases} 1 & \text{for } |\omega| \leq \frac{\pi}{2} \\ 0 & \text{for } \frac{\pi}{2} \leq |\omega| \leq \pi \end{cases}$$

Find the frequency response and the unit impulse response of the system. (1+1)

4. Consider the linear constant-coefficient difference equation

$$y[n] - \frac{1}{4}y[n-1] - \frac{1}{8}y[n-2] = 2x[n-1]$$

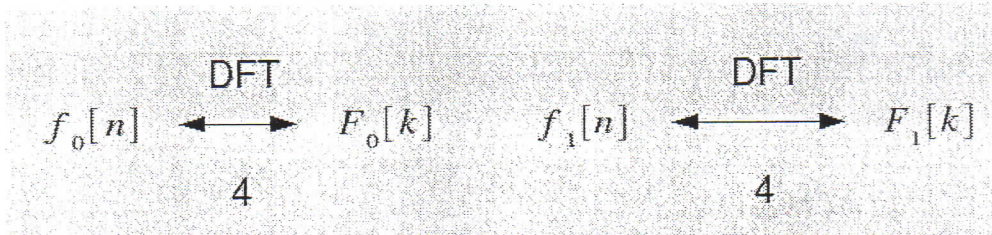
Determine $y[n]$ for $n \geq 0$ when $x[n] = \delta[n]$ and $y[n] = 0, n < 0$ (1.5)

5. As part of the first stage in a radix 2 FFT, a sequence $x[n]$ of length $N=8$ is decomposed into 2 sequences of length 4 as:

$$f_0[n] = x[2n], n=0,1,2,3$$

$$f_1[n] = x[2n+1], n=0,1,2,3$$

We compute a 4 point DFT of each of these 2 sequences as



The specific values of $F_0[k]$ and $F_1[k]$ ($k=0,1,2,3$), obtained from the length $N=8$ sequence in question are listed in the table below:

k	0	1	2	3
$F_0[k]$	0	0	1	1
$F_1[k]$	0	0	-i	$\frac{1}{\sqrt{2}}(-1+i)$
W_8^k	1	$\frac{1}{\sqrt{2}}(1-i)$	-i	$\frac{1}{\sqrt{2}}(1+i)$

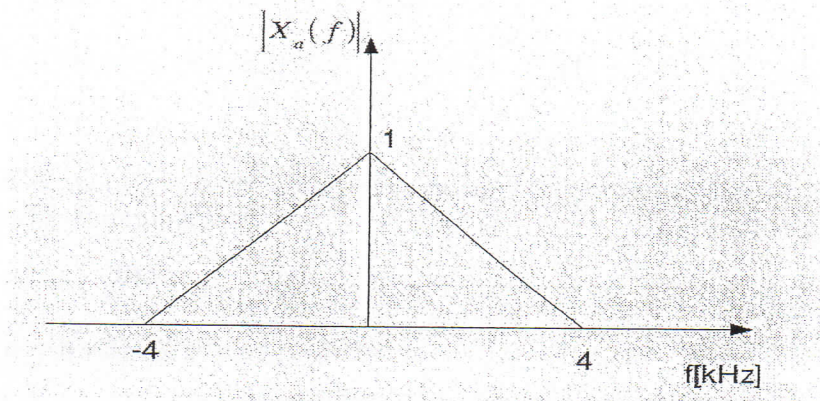
(a) From the values of $F_0[k]$ and $F_1[k]$ ($k=0,1,2,3$) and the values of W_8^k ($k=0,1,2,3$), provided in the table, determine the numerical value of the actual $N=8$ point DFT of $x[n]$ denoted $X_s[k]$. That is, determine the numerical value of $X_s[k]$ for $k=0,1,\dots,7$. (1.5)

(b) The underlying length $N=8$ sequence $x[n]$ may be expressed as

$$x[n] = \frac{1}{4} e^{2\pi i \frac{k_1}{8} n} + \frac{1}{4} e^{2\pi i \frac{k_2}{8} n} \text{ where } k_1 \text{ and } k_2 \text{ are both integers between 0 and 7. Given the}$$

values of $X_s[k]$ determined in part (a) determine the numerical values of k_1 and k_2 . (0.5)

6. Let $x_1(t)$ be an analog signal, with an amplitude spectrum $(X_a(f))$ shown in the following figure:



A discrete-time signal $x_1(n)$ was generated by sampling the signal $x_a(t)$ using the sampling frequency $F_{s1} = 8 \text{ kHz}$.

We wish to design a digital system that reduces the sampling frequency of the signal $x_1(n)$ to $F_{s2} = 6 \text{ kHz}$ such that aliasing does not appear. Let $x_2(m)$ be the resulting output signal.

- Sketch the block scheme of the system (including up-samplers, down-samplers, filters, etc.) and explain the function of each component. State the necessary specifications of the components of the system. **(1)**
- Will the reduction in the sampling frequency cause a loss of information, i.e. is it possible to reconstruct the analog signal $x_a(t)$ from the output signal $x_2(m)$? **(0.5)**
- Sketch the magnitude spectra of all the signals in the system (i.e. $(X_1(f))$, $(X_2(f))$ and possible intermediate signals) for frequencies up to at least $2x$ the sampling frequency. State the value of the corresponding sampling frequency for each graph. **(0.5)**

$$\begin{aligned}
 \textcircled{1} \quad X(e^{j\omega}) &= \sum_{n=0}^{\infty} \left(\frac{1}{3}\right)^n e^{-j\omega n} + \sum_{n=-\infty}^0 \left(\frac{1}{3}\right)^{-n} e^{-j\omega n} - 1 \\
 &= \frac{1}{1 - \frac{1}{3}e^{-j\omega}} + \sum_{n=0}^{\infty} \left(\frac{1}{3}\right)^n e^{j\omega n} - 1 \\
 &= \frac{1}{1 - \frac{1}{3}e^{-j\omega}} + \frac{1}{1 - \frac{1}{3}e^{j\omega}} - 1 = \frac{8/9}{1/9 - \frac{2}{3}\cos\omega}
 \end{aligned}$$

$$\textcircled{2} \text{ (a)} \quad x[n] = \left(\frac{1}{3}\right)^n \mu[n]$$

$$X(z) = \sum_{n=-\infty}^0 \left(\frac{1}{3}\right)^n z^{-n} = \sum_{n=0}^{\infty} \left(\frac{1}{3}\right)^n z^n = \frac{1}{1-3z}$$

$$\textcircled{b} \quad x[n] = \left(\frac{1}{2}\right)^n \mu[-n+2] + 3^n \mu[n-1]$$

$$X(z) = \sum_{n=-\infty}^2 \left(\frac{1}{2}\right)^n z^{-n} + \sum_{n=1}^{\infty} 3^n z^{-n}$$

$$= \sum_{n=-2}^{\infty} \left(\frac{1}{2}\right)^{-n} z^n + \sum_{n=1}^{\infty} 3^n z^{-n}$$

$$= \left(\frac{1}{2}\right)^{-2} z^{-2} \left(\sum_{n=0}^{\infty} \left(\frac{1}{2}\right)^{-n} z^n \right) + 3z^{-1} \sum_{n=0}^{\infty} 3^n z^{-n}$$

$$= \frac{1}{4z^2} \cdot \frac{1}{1-2z} + \frac{3}{z} \cdot \frac{1}{1-\frac{3}{z}} =$$

$$= \frac{1}{4z^2(1-2z)} + \frac{3}{z-3} \quad (\text{can be simplified more}).$$

$$2(c) \quad x[n] = 2n \left(\frac{1}{2}\right)^n \mu[n+1]$$

$$\text{Use here: } \mathcal{Z}(ng[n]) = -z \frac{dG(z)}{dz}$$

$$\text{Let } g[n] = 2 \cdot \left(\frac{1}{2}\right)^n \mu[n+1]$$

$$G(z) = 2 \sum_{n=-1}^{\infty} \left(\frac{1}{2}\right)^n z^{-n} = 2 \cdot \left(\frac{1}{2}\right)^{-1} z \left(\sum_{n=0}^{\infty} \left(\frac{1}{2}\right)^n z^{-n} \right)$$

$$= 4z \frac{1}{1 - \frac{1}{2z}} = \frac{8z^2}{2z - 1}$$

$$\frac{dG(z)}{dz} = \frac{16z \cdot (2z - 1) - 8z^2 \cdot 2}{(2z - 1)^2} = \frac{32z^2 - 16z - 16z^2}{(2z - 1)^2}$$

$$= \frac{16(z^2 - z)}{(2z - 1)^2}$$

$$\text{So } \mathcal{Z}(x[n]) = -z \left(\frac{16(z^2 - z)}{(2z - 1)^2} \right) = \frac{-16z^3 + 16z^2}{(2z - 1)^2}$$

③

To find the unit sample response, let $x[n] = \delta[n]$.

The output of the adder is

$$w[n] = \delta[n] + \delta[n-1]$$

So

$$y[n] = h_2[n] + h_2[n-1]$$

Here

$$h_2[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} H_2(e^{j\omega}) e^{jn\omega} d\omega = \frac{1}{2\pi} \int_{-\pi/2}^{\pi/2} e^{jn\omega} d\omega$$

$$= \frac{\sin(n\pi/2)}{n\pi}$$

So, the unit impulse response of the ^{whole} system is:

$$h[n] = \frac{\sin(n\pi/2)}{n\pi} + \frac{\sin((n-1)\pi/2)}{(n-1)\pi}$$

For the frequency response:

$$W(e^{j\omega}) = 1 + e^{-j\omega}$$

$$\text{So } H(e^{j\omega}) = W(e^{j\omega})H_2(e^{j\omega}) = (1 + e^{-j\omega})H_2(e^{j\omega})$$

$$= \begin{cases} 1 + e^{-j\omega} & (|\omega| \leq \pi/2) \\ 0 & (\pi/2 < |\omega| \leq \pi) \end{cases}$$

(4)

$$y[n] = -\frac{1}{4}y[n-1] - \frac{1}{8}y[n-2] = 2x[n-1]$$

When $x[n] = \delta[n]$, $y[n] = h[n]$ (impulse response)

We calculate $H(e^{j\omega})$ from:

$$Y(e^{j\omega}) - \frac{1}{4}Y(e^{j\omega})e^{-j\omega} - \frac{1}{8}Y(e^{j\omega})e^{-2j\omega} = 2X(e^{j\omega})e^{-j\omega}$$

$$\text{So } H(e^{j\omega}) = \frac{Y(e^{j\omega})}{X(e^{j\omega})} = \frac{2e^{-j\omega}}{1 - \frac{1}{4}e^{-j\omega} - \frac{1}{8}e^{-2j\omega}}$$

$$= \frac{A}{1 - \frac{1}{2}e^{-j\omega}} + \frac{B}{1 + \frac{1}{4}e^{-j\omega}} = \frac{A(1 + \frac{1}{4}e^{-j\omega}) + B(1 - \frac{1}{2}e^{-j\omega})}{(1 - \frac{1}{2}e^{-j\omega})(1 + \frac{1}{4}e^{-j\omega})}$$

In that case

$$\begin{cases} A + B = 0 \\ +\frac{1}{4}A - \frac{1}{2}B = 2 \end{cases} \quad \text{or} \quad \begin{cases} A + B = 0 \\ -\frac{1}{4}B - \frac{1}{2}B = 2 \end{cases}$$

$$(B = -\frac{8}{3}, A = \frac{8}{3})$$

$$= \frac{8/3}{1 - \frac{1}{2}e^{-j\omega}} - \frac{8/3}{1 + \frac{1}{4}e^{-j\omega}}$$

$$\text{So } h[n] = \frac{8}{3} \left(\frac{1}{2}\right)^n \mu[n] - \frac{8}{3} \left(\frac{1}{4}\right)^n \mu[n] = y[n]$$

⑤ Here we use (see derivation of FFT):

$$X_s[k] = F_0[k] + W_8^k F_1[k] \quad (k=0,1,2,3)$$

$$X_s[k+4] = F_0[k] - W_8^k F_1[k]$$

This gives:

$$X_s[0] = 0$$

$$X_s[4] = 0$$

$$X_s[1] = 0$$

$$X_s[5] = 0$$

$$X_s[2] = 1 + (-i)(-i) = 0$$

$$X_s[6] = 2$$

$$X_s[3] = 2$$

$$X_s[7] = 0$$

$$\text{So } X_s[k] = 2\delta[k-3] + 2\delta[k-6].$$

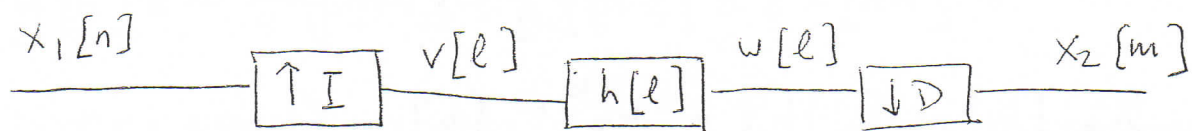
Since we know that the IDFT of $\delta(k-k_0) = \frac{1}{N} W_N^{-k_0}$

we have:

$$x[n] = \frac{1}{4} e^{2\pi i/8 \cdot 3} + \frac{1}{4} e^{2\pi i/8 \cdot 6}$$

$$\text{So } k_1 = 3, k_2 = 6.$$

6



(Block diagram)

(a)

$\boxed{\uparrow I}$

Interpolator, inserts $I-1$ zero-samples between the samples of $x_1[n]$. Here the sampling frequency should go up by a factor 3 ($=I$)

$\boxed{h[l]}$

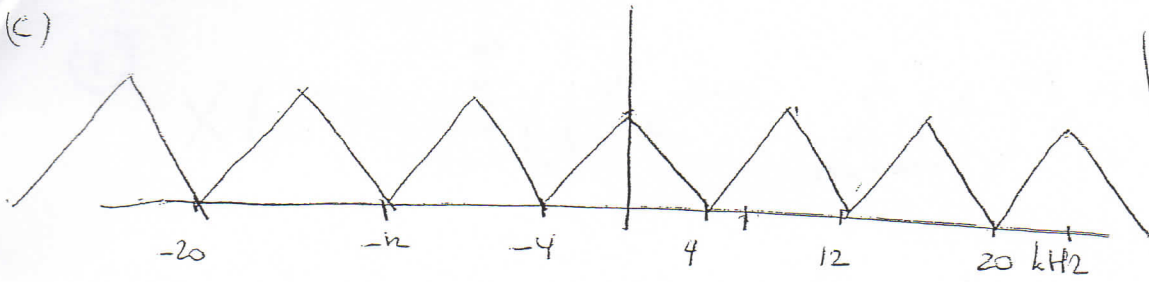
Digital low-pass filter, that removes all ~~flux~~ above frequencies above $F_c = 3$ kHz. Operates at sampling frequency 24 kHz.

$\boxed{\downarrow D}$

Decimator, retaining only each D^{th} sample. In this case $D=4$.

(b) Since the bandwidth of the signal is larger than $F_{S_2} = 6$ kHz, it is not possible to reconstruct $x_a(t)$ from $x_2[m]$. So, some information is lost.

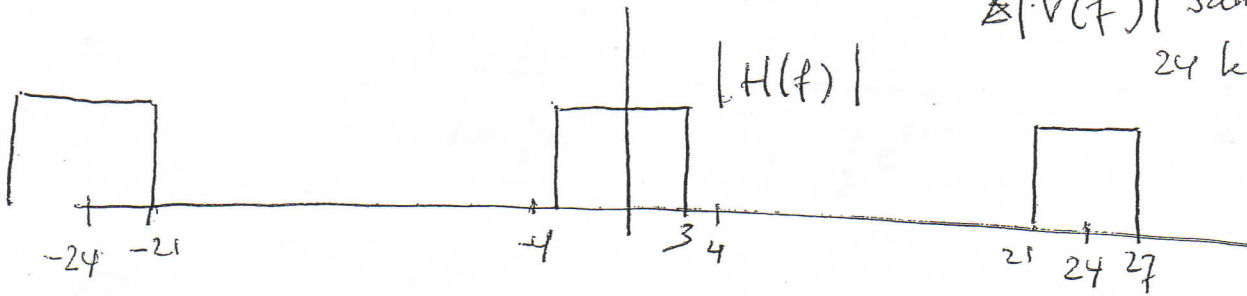
(c)



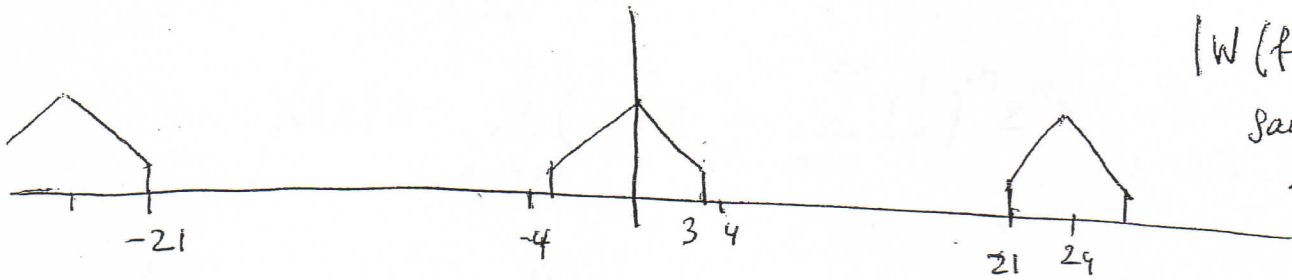
$|X_1(f)|$
Sampled at
8 kHz

Same for

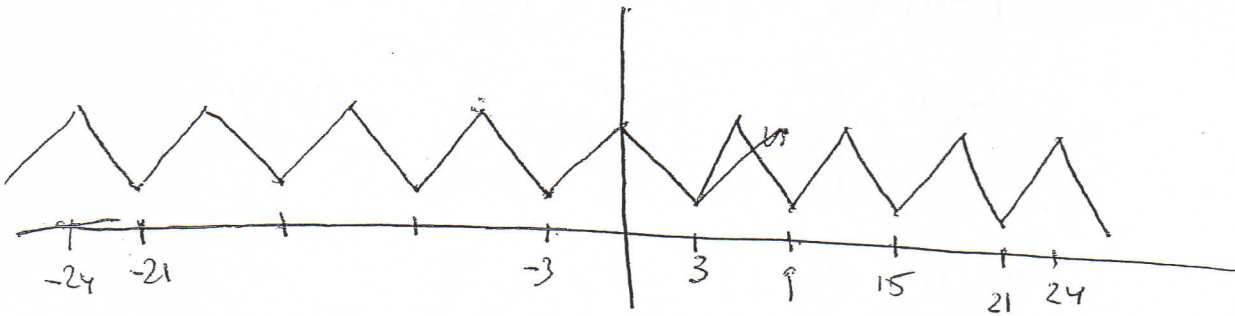
$|V(f)|$ sampled at
24 kHz.



$|H(f)|$



$|W(f)|$
sampled at
24 kHz



$|X_2(f)|$
sampled at
6 kHz