$$7. (a) \quad X[n] = 2\cos(\frac{\pi q n}{3}) = e^{j(\frac{\pi q n}{3})} + e^{-j(\frac{\pi q n}{3})}$$

$$X(w) = \sum_{n=-\infty}^{\infty} X[n] e^{jwn} = \sum_{n=-\infty}^{\infty} (e^{j(\frac{\pi q}{3})} + e^{-j(\frac{\pi q}{3})}) e^{jwn}$$

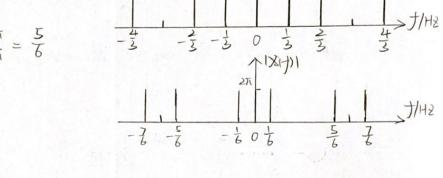
$$= \sum_{n=-\infty}^{\infty} (e^{j(\frac{\pi q}{3} - w)n} + e^{-j(\frac{\pi q}{3} + w)n})$$

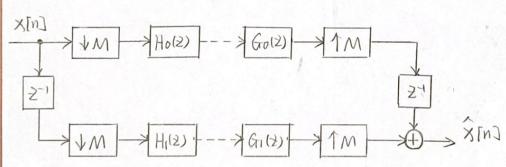
$$= 2\pi \sum_{k=-\infty}^{\infty} [8(w - \frac{\pi q}{3} - 2\pi k) + 8(w + \frac{\pi q}{3} - 2\pi k)]$$

$$9 = 1 : \quad \int_{1}^{\infty} \frac{w}{2\pi} = \frac{\pi q}{2\pi} = \frac{1}{3}$$

$$9 = 2 : \quad \int_{2}^{\infty} \frac{w}{2\pi} = \frac{3\pi}{2\pi} = \frac{1}{3}$$

$$9 = 5 : \quad \int_{3}^{\infty} \frac{w}{2\pi} = \frac{3\pi}{2\pi} = \frac{5\pi}{3}$$





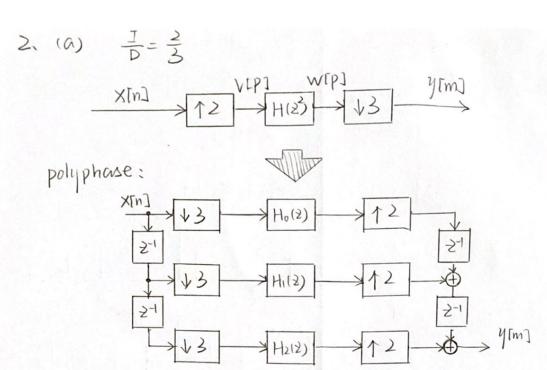
$$H_0(2) = P_0(2^2) + 2^{-1}P_1(2^2)$$

$$H_1(2) = P_0(2^2) - 2^{-1}P_1(2^2)$$

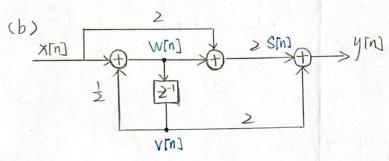
$$G_0(2) = P_0(2^2) + 2^{-1}P_1(2^2)$$

$$G_1(2) = -P_0(2^2) + 2^{-1}P_1(2^2)$$

(b)



The subfilters are norking at the lowest sampling trequency.



$$W(2) = X(2) + \frac{1}{2}V(2), \quad V(2) = 2^{-1}W(2),$$

$$= > W(2) = X(2) + \frac{1}{2}2^{-1}W(2), \quad X(2) = (1 - \frac{1}{2}2^{-1})W(2)$$

$$S(2) = W(2) + 2X(2)$$

$$Y(2) = 2V(2) + 2S(2) = 22^{-1}W(2) + 2W(2) + 4X(2)$$

$$= (22^{-1} + 2)W(2) + (4 - 22^{-1})W(2)$$

$$= 6W(2)$$

$$\frac{Y(2)}{X(2)} = \frac{6}{1 - \frac{1}{2}z^{-1}}$$

$$cu3 u_0 t = \frac{1}{2} (e^{jw_0 t} + e^{-jw_0 t}), \quad sin w_0 t = \frac{1}{2j} (e^{jw_0 t} - e^{-jw_0 t})$$

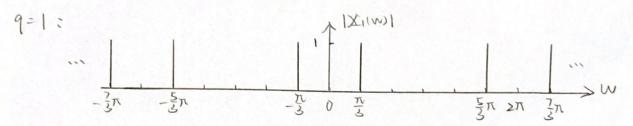
$$\Rightarrow P[cu3 u_0 t] = \pi[S(u_1 + w_0) + S(w_1 - w_0)]$$

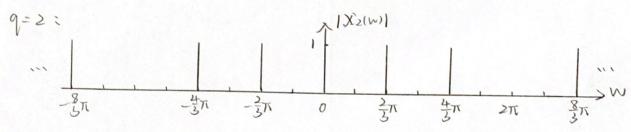
$$F[sin w_0 t] = j\pi[S(w_1 + w_0) - S(w_1 - w_0)]$$

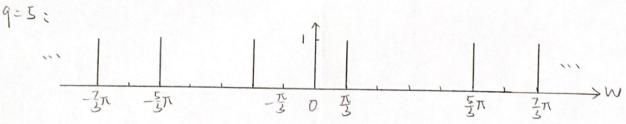
$$X[n] = 2\cos(\frac{1}{5}\pi qn) = e^{\frac{1}{3}(\frac{1}{5}\pi qn)} + e^{-\frac{1}{3}(\frac{1}{5}\pi qn)}.$$

$$X(w) = \sum_{n=-\infty}^{\infty} x[n]e^{-\frac{1}{3}un} = \sum_{n=-\infty}^{\infty} (e^{\frac{1}{3}(\frac{1}{3}\pi q-w)n} + e^{-\frac{1}{3}(\frac{1}{3}\pi q+w)n})$$

$$= 2\pi \sum_{k=-\infty}^{\infty} [\delta(w+\frac{1}{3}\pi q-2k\pi) + \delta(w-\frac{1}{3}\pi q-2k\pi)]$$







$$F[S(+)] = \int_{-\infty}^{+\infty} S(+) e^{-jwt} dt = e^{-jwt} \Big|_{0} = 1,$$

$$F^{-1}[1(+)] = \frac{1}{2\pi} \int_{-\infty}^{+\infty} 1 \cdot e^{jwt} dw = S(+),$$

$$= \sum_{-\infty}^{+\infty} e^{jwt} dw = 2\pi S(+)$$

3. (a) parameters used to specify LPF:

O passband treq wp = 27 Ft.

@ stopband freq ws = 21 = 1

3 passband ripple $\delta p = 0.5(1-10^{\frac{-\alpha p}{20}})$,

4 stopband ripple 8s = 10-20

(3) filter order $N = \frac{-10 \log_{10} (SpSs) - 13}{14.6 \times (ws - w_p)/2\pi}$

SpSs $\downarrow \rightarrow N\uparrow$; N is inversly proportional to the size of ripples.

(ws-wp) $\downarrow \rightarrow N\uparrow$;
N is inversly proportional to the transition bandwidth.

Nis proportional to the sampling freq.

Stage 1: Pi=3, Spi= \frac{1}{2}Sp=0.005, Ssi=Ss=0.001.

passband treq: 0~450HZ

stopband treq: 450 ~ (48K/3)-450 = 15550 HZ,

 $\Delta F_1 = (15550 - 450) / 48 \times 10^3 = 0.31458$

$$N_{1} = \frac{-10 \log_{10} (Sp_{1}Ss_{1}) - 13}{14.6 \times \Delta F_{1}} + 1 = 9.7 \approx 10$$

Stage 2: Dz=16, Sp= = = Sp= a005, Ssz= Ss= a001

passband treq: on 450Hz

Stopband treq: 450~500 Hz.

input sampling treq: $FT = F_{7}/3 = 10 \times 10^{3} Hz$, $F_{5} = \frac{F_{7}}{2p} = \frac{16 \times 10^{3}}{32}$ $\Delta F = (500 - 450)/16 \times 10^{3} = 0.003125$, = 500 Hz,

$$N_2 = \frac{-10\log_{10}(Ss_2Sp_2) - 13}{14.6 \times Df_2} + 1 = 877.9 \approx 878$$

 $N_1=10$, $N_2=878$ $\frac{(N+1)}{2} \times F_1$ multiplications/sec $N \times F_1$ additions/sec

 \Rightarrow Stage 1: $\frac{10+1}{2} \times 48 \times 10^3 = 264 \times 10^3$ multiplications/sec : $10 \times 48 \times 10^3 = 48 \times 10^4$ additions/sec ;

Stage 2: $\frac{878+1}{2} \times 16 \times 10^3 = 7032 \times 10^3 = 7.032 \times 10^6 \text{ multiplications/sec}$ $878 \times 16 \times 10^3 = 1.4048 \times 10^7 \text{ additions/sec}$

4
$$\times [n] = S[n] + w[n]$$
, $S[n] - \alpha 4S[n-1] = v[n]$, $T_{v}^{2} = \alpha 56$

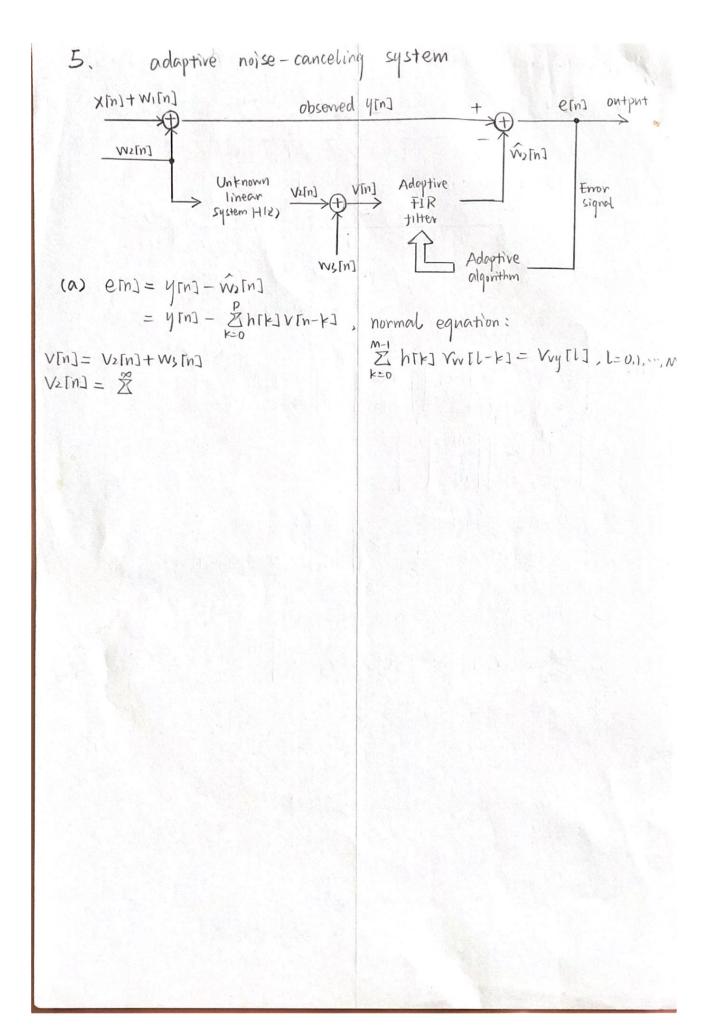
$$= \frac{\alpha 56}{(1 - \alpha 42^{-1})(1 - \alpha 42)} = \frac{\alpha 36}{\alpha 84} \cdot \frac{1}{1 - \alpha 42^{-1}}$$

$$= \frac{(1 - \alpha 42^{-1})(1 - \alpha 42)}{(1 - \alpha 42) + \alpha 56} = \frac{1.52 - \alpha 4(2^{-1} + 2)}{(1 - \alpha 42^{-1})(1 - \alpha 42)}$$

$$= \frac{(1 - \alpha 42^{-1})(1 - \alpha 42)}{(1 - \alpha 42^{-1})(1 - \alpha 42)} = \frac{1.52 - \alpha 4(2^{-1} + 2)}{(1 - \alpha 42^{-1})(1 - \alpha 42)}$$

$$= \frac{1}{2} \times (\alpha 4)^{|K|}, \quad \text{Yxx} [K] = \text{Vax} [K] + \text{Vxx} [K]$$

$$= \frac{1}{2} (\alpha 4)^{|K|} + S[K]$$



4. (a)
$$\lceil \csc(2) = (\sqrt{1+|2^{-1}|+|2^{-1}|}) + (2) \rceil$$

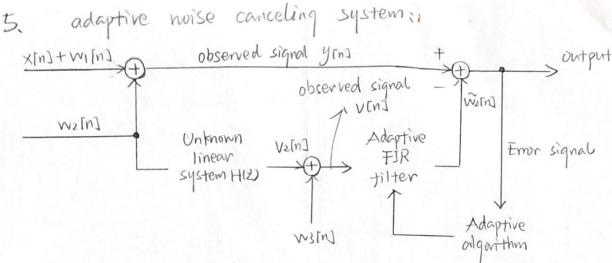
$$= \frac{0.36}{(1-0.42^{-1})(1-0.42)} = \frac{0.36}{0.84} \frac{1}{1-0.42^{-1}} = \frac{3}{7} \frac{1}{1-0.42^{-1}},$$

$$YSS[m] = \frac{3}{7} (0.4)^{[m]}, \quad Vxx[m] = VSS[m] + Vxx[m]$$

$$= \frac{3}{7} (0.4)^{[m]} + S[m],$$

(b)
$$\begin{bmatrix} w_{1} & w_{1} & w_{1} \\ w_{1} & w_{1} & w_{1} \end{bmatrix} \begin{bmatrix} h_{1} & w_{1} \\ h_{1} & w_{2} \end{bmatrix} = \begin{bmatrix} v_{35} & v_{1} \\ v_{55} & w_{1} \end{bmatrix}$$
.
 $\begin{bmatrix} v_{1} & v_{2} & w_{1} \\ v_{2} & w_{2} & w_{2} \end{bmatrix} \begin{bmatrix} h_{1} & w_{1} \\ h_{1} & w_{2} \end{bmatrix} = \begin{bmatrix} v_{1} & v_{2} \\ v_{2} & w_{2} \\ v_{2} & w_{2} \end{bmatrix}$.
 $\begin{bmatrix} h_{1} & w_{1} \\ h_{1} & w_{2} \end{bmatrix} = \begin{bmatrix} \alpha_{1} & \alpha_{2} & \alpha_{3} \\ \alpha_{2} & w_{3} & w_{2} \end{bmatrix} \begin{bmatrix} v_{2} & w_{3} \\ h_{1} & w_{2} \end{bmatrix} \begin{bmatrix} v_{2} & w_{3} \\ h_{1} & w_{2} \end{bmatrix}$

$$= \begin{bmatrix} \alpha_{2} & w_{3} & w_{3} \\ \alpha_{2} & w_{3} & w_{3} \end{bmatrix} \begin{bmatrix} h_{1} & w_{2} & w_{3} \\ h_{1} & w_{3} & w_{3} \end{bmatrix} \begin{bmatrix} h_{1} & w_{2} & w_{3} \\ h_{1} & w_{3} & w_{3} \end{bmatrix} = 0.2898$$
.



X[n] is a stationary random process, $V_1[n]$, $V_2[n]$, $V_3[n]$ are white noise sequence with $T_{w_1} = a_0 4$, $T_{w_2} = a_1 b$, $T_{w_3} = 1$. $V_1[n] = V_2[n] + W_3[n]$, $V_1[n] = X[n] + W_1[n] + W_2[n]$; $V_2[n] = V_2[n] = V_1[n] - \sum_{k=0}^{M-1} h_{[k]} V_{[n-k]}$; $V_2[n] = \sum_{k=0}^{M-1} h_{[k]} V_{[n-k]}$; normal equation: $V_{w_1}[l] = \sum_{k=0}^{M-1} h_{[k]} V_{w_1}[n-l]$. $V_{w_2}[l] = \sum_{k=0}^{M-1} h_{[k]} V_{w_2}[n-l]$.

 $= \frac{1}{N} \sum_{n=0}^{N-1} \left(\sum_{l=0}^{\infty} h[l] w_{2}[n-l] + w_{3}[n] \right) \left(\sum_{p=0}^{\infty} h[p] w_{2}[n-k-p] + w_{3}[n-k] \right)$

= 1 2 2 hrpjh[l] W2[n-k-p] W2[n-l] + W3[n] 2 hrpjw2[n-k-p] +

W3[n-K] & h[l] W2[n-t] + W3[n] W3[n-k])

E[VIVEK]] = Ywsws[k] + Zhel] Ywzws[k-1] + Zhep] Vwsws[k+p] + Zhep hep hep hell Ywzws[k+p-1].

Vy[k] = 1 2 ym] v[n-k] = 1 2 (x[n]+w,[n]+w2[n])(2 h[] w2[n-k-l]+ w2[n-l])

= 1 2 2 2 (X[n]+w,[n]+w,[n]) h[1] w2 [n-k-1] + 1 2 (X[n]+w,[n]+w2[n]).

E[Vry[k]= = h[l] (Xxw2[k+l] + Yww, [k+l] + Vw2w2 [k+l]) + Vxw, [l] + Vw2w, [l]
+ Vw2w, [l]

[Inhm=Yy], $[Infn+1]=[Imfn]+\frac{1}{2}\Delta[n]S[n]=[Imfn]-\frac{1}{2}\Delta[n]g[n]$ $g[n]=\frac{dEm}{dhm}=2([Imhm-Yy])$

EE6401

NANYANG TECHNOLOGICAL UNIVERSITY SEMESTER 1 EXAMINATION 2017-2018 EE6401 – ADVANCED DIGITAL SIGNAL PROCESSING

November/December 2017

Time Allowed: 3 hours

INSTRUCTIONS

- 1. This paper contains 5 questions and comprises 3 pages.
- 2. Answer all 5 questions.
- 3. All questions carry equal marks.

4. This is a closed-book examination.

 $X(w) = 2\pi \sum_{k=-\infty}^{\infty} \left(S(w + \frac{9\pi}{3}\pi - 2k\pi)\right) + S(w - \frac{9\pi}{3}\pi - 2k\pi)$

1. (a) Find the discrete-time Fourier transform (DTFT) of the signal

$$x[n] = 2\cos(\frac{\pi qn}{3})$$

for q = 1, 2 and 5. Plot these DTFTs and comment on your observations.

(8 Marks)

(b) For a two-channel alias-free quadrature filter bank, the filters $H_0(z)$, $H_1(z)$, $G_0(z)$ and $G_1(z)$ in the analysis and synthesis sections can be implemented with a two-branch polyphase filter structure. For example, $H_0(z) = P_0(z^2) + z^{-1}P_1(z^2)$, where $P_0(z)$ and $P_1(z)$ are the expressions of the branch filters in the polyphase implementation. Find a computationally efficient polyphase implementation of the two-channel alias-free quadrature filter bank.

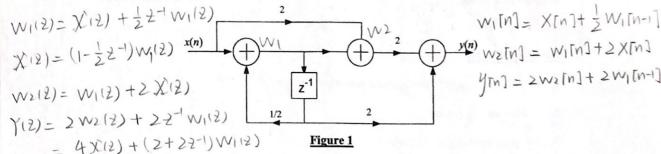
(12 Marks)

A sampling rate converter is needed to change its input sampling frequency by 2/3 times. To minimize the computational costs, derive a polyphase structure in its most efficient form, and explain why the computational complexity is reduced by the polyphase implementation compared to the implementation directly using a finite impulse response (FIR) filter.

(14 Marks)

Determine the z-transform of the impulse response of the system shown in Figure 1

(6 Marks)



(a) List the parameters that are used to specify a low-pass digital filter and generate the filter coefficients. Briefly discuss how these parameters affect the filter order. $= (4(1-\frac{1}{2}z^{-1}) + 2 + 2z^{-1}) W_1(2)$

$$= \frac{1 - \frac{7}{5} + 1}{6} \times 15$$

The sampling rate of a signal x[n] is to be reduced, by decimation, from 48 kHz to 1 kHz. The highest frequency of interest after decimation is 450 Hz. Assume that = $\frac{1 \text{ kHz}}{1 - \frac{1}{2} \cdot 1}$ X is an optimal FIR filter is to be used, with an overall passband ripple, $\delta_p = 0.01$, and stopband ripple, $\delta_r = 0.001$. Design an efficient decimator with two-stage FIR filters shown in Figure 2, and calculate the required computational complexity in terms of the number of additions and multiplications per second.

Stage 1:
$$PF = \frac{16}{3} + PF = \frac{16}{2F_{SL}}$$
 $F_{T1} = \frac{48}{8} \times \frac{1}{3} = \frac{16}{16} \times \frac{16}{2F_{SL}}$
 $F_{SL} = \frac{16}{16} \times \frac{45}{16} = \frac{15}{15} \times \frac{16}{16} \times \frac{16}{1$

$$T_{SS(2)} = T_{V}^{2} + (2) + (2) = \frac{\alpha 36}{(1 - \alpha 42^{-1})(1 - \alpha 42)} = \frac{(1 - \alpha 42^{-1})(1 - \alpha 42)}{(1 - \alpha 42^{-1})(1 - \alpha 42)} = \frac{(1 - \alpha 42^{-1})(1 - \alpha 42)}{(1 - \alpha 42^{-1})(1 - \alpha 42)}$$

$$EE6401$$

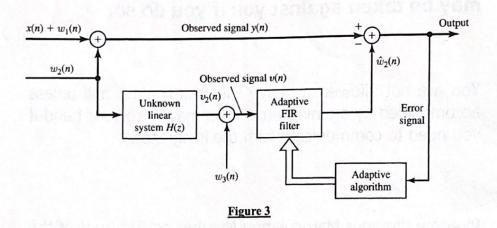
- 4. Consider a signal x[n] = s[n] + w[n] where $\{s[n]\}$ is an AR(1) process that satisfies the difference equation s[n] = 0.4s[n-1] + v[n] where w[n] and v[n] are uncorrelated white noise sequences with variances $\sigma_v^2 = 0.36, \sigma_w^2 = 1$.
 - (a) Determine the power density spectrum of x[n].

(10 Marks)

(b) Design a Weiner filter of length M = 2 to estimate s[n].

(10 Marks)

5. Consider an adaptive noise-canceling system illustrated in Figure 3 where x[n] is a stationary random process and $w_1[n], w_2[n], w_3[n]$ are white noise sequences with variances $\sigma_{w_1}^2 = 0.04, \sigma_{w_2}^2 = 0.16, \sigma_{w_3}^2 = 1$, respectively.



(a) Determine the equations of the overall system.

(10 Marks)

(b) Explain how the Least Mean Square (LMS) Algorithm can be applied in this application.

(10 Marks)

END OF PAPER