

Chap12 Multirate Signal Processing

Chao-Tsung Huang

National Tsing Hua University Department of Electrical Engineering

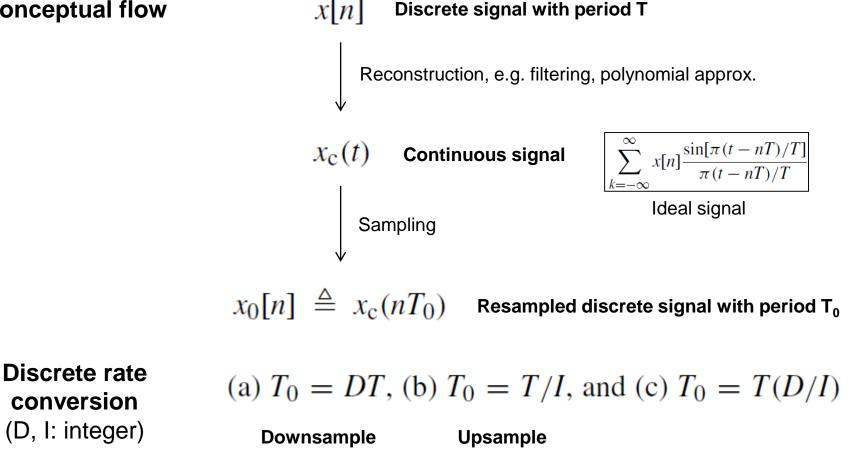


Chap 12 Multirate signal processing

- 12.1 Sampling rate conversion
- 12.2 Implementation of multirate systems
- 12.3 Filter design for multirate systems
- 12.4 Two-channel filter banks

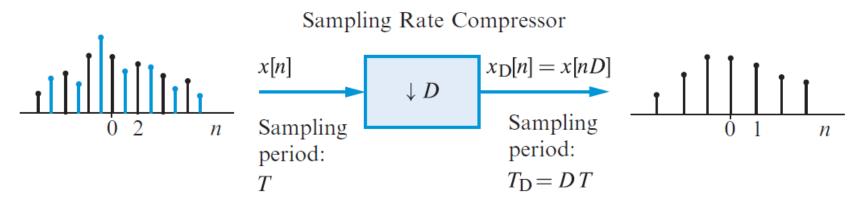
Sampling rate conversion (Resampling

Conceptual flow



Downsampler: sampling rate compressor





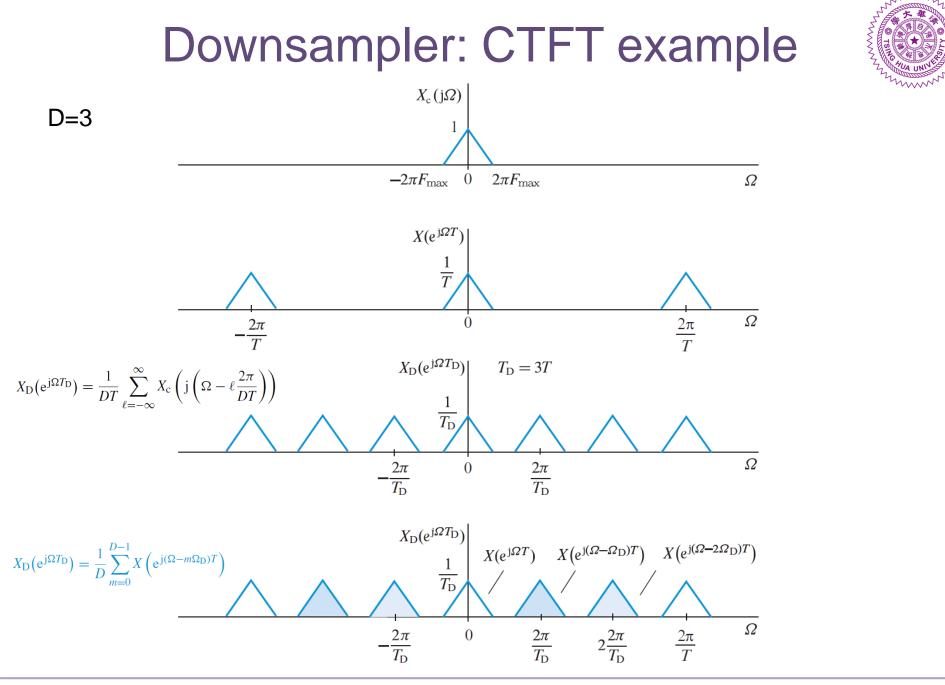
$$X_{\rm D}({\rm e}^{{\rm j}\Omega T_{\rm D}}) = \frac{1}{DT} \sum_{\ell=-\infty}^{\infty} X_{\rm c}\left({\rm j}\left(\Omega - \ell \frac{2\pi}{DT}\right)\right)$$

$$X_{\rm D}(\mathrm{e}^{\mathrm{j}\Omega T_{\rm D}}) = \frac{1}{D} \sum_{m=0}^{D-1} X\left(\mathrm{e}^{\mathrm{j}(\Omega - m\Omega_{\rm D})T}\right)$$

$$X_{\rm D}({\rm e}^{{\rm j}\omega}) = \frac{1}{D} \sum_{m=0}^{D-1} X\left({\rm e}^{{\rm j}(\omega-2\pi m)/D}\right) \qquad \omega \triangleq \Omega T_{\rm D}$$

DTFT

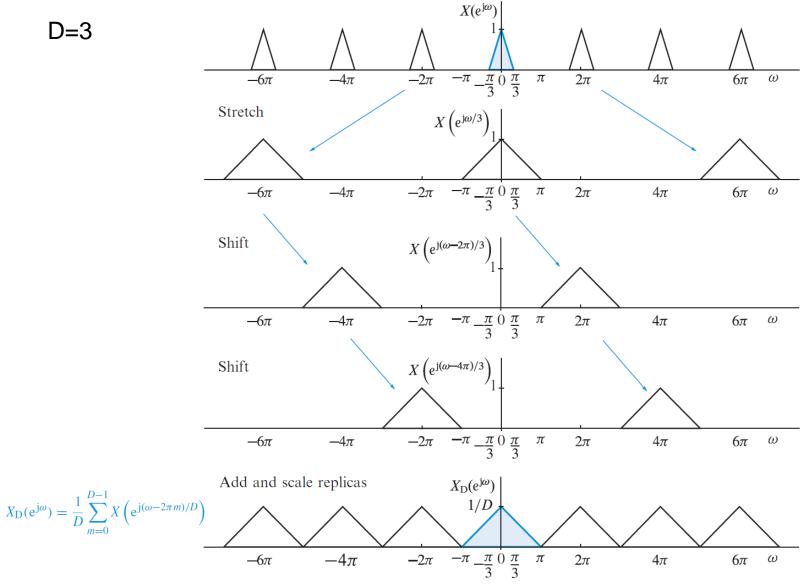
CTFT



Downsampler: DTFT example







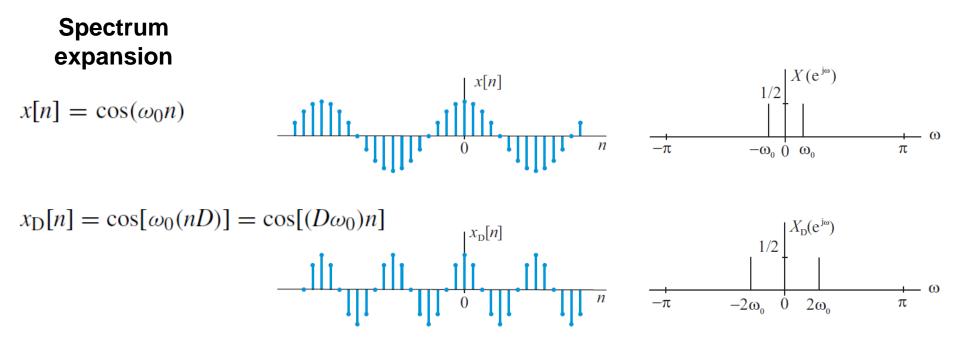
Downsampler: D=2 example



General case

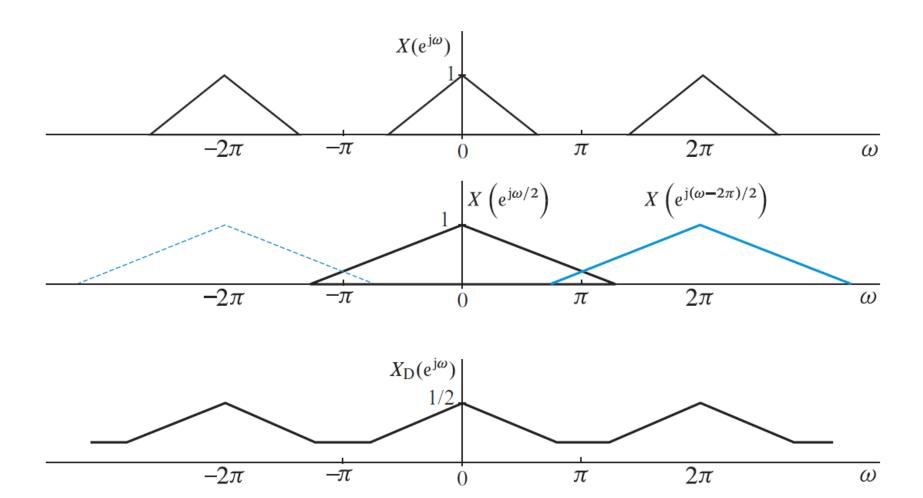
$$x[n] = \{\dots, x[-2], x[-1], x[0], x[1], x[2], \dots\}$$
$$\mathcal{D}_2\{x[n]\} = \{\dots, x[-4], x[-2], x[0], x[2], x[4], \dots\} = x_D[n]$$
$$\mathcal{D}_2\{x[n+1]\} = \{\dots, x[-3], x[-1], x[1], x[3], x[5], \dots\}$$

$$X_{\rm D}({\rm e}^{{\rm j}\omega}) = \frac{1}{2}X({\rm e}^{{\rm j}\omega/2}) + \frac{1}{2}X({\rm e}^{{\rm j}(\omega/2-\pi)})$$

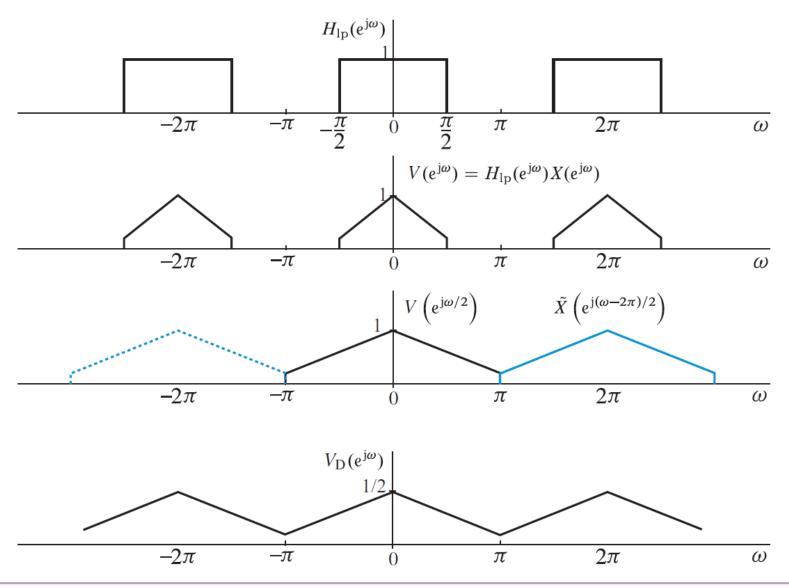


Downsampling with aliasing (D=2)



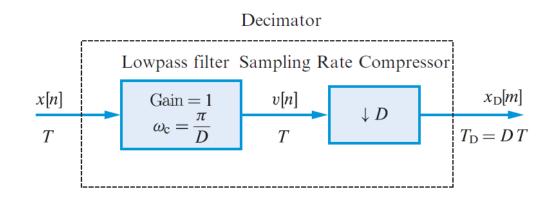


Downsampling with lowpass prefiltering (Decimation)



Decimator





For FIR filter

$$x_{\rm D}[m] = v[mD] = \sum_{k=0}^{M} h[k]x[mD-k]$$

. .

Overall computation can be reduced to 1/D

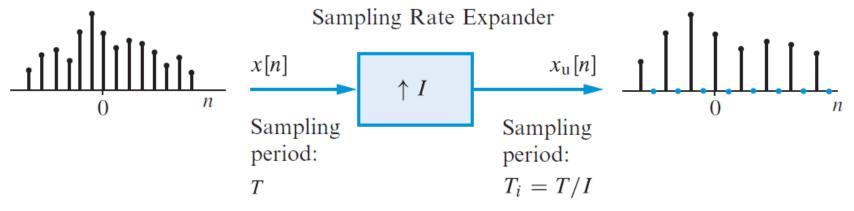
To avoid aliasing

If
$$X(e^{j\omega}) = 0$$
, $\omega_{\rm H} \le |\omega| \le \pi$ then $\omega_{\rm s} = \frac{2\pi}{D} \ge 2\omega_{\rm H}$

Safe choice $\omega_c = \pi/D$

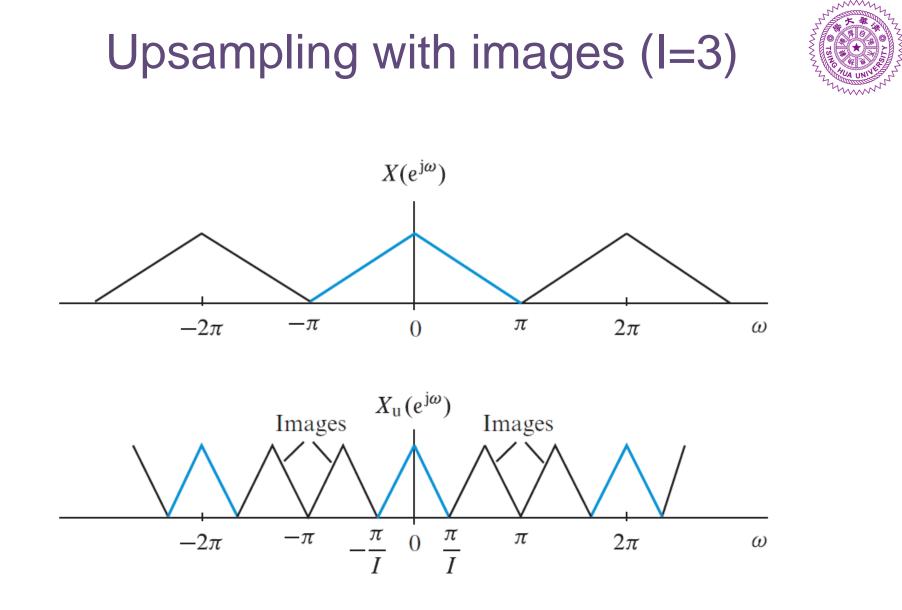
Upsampler: sampling rate expander





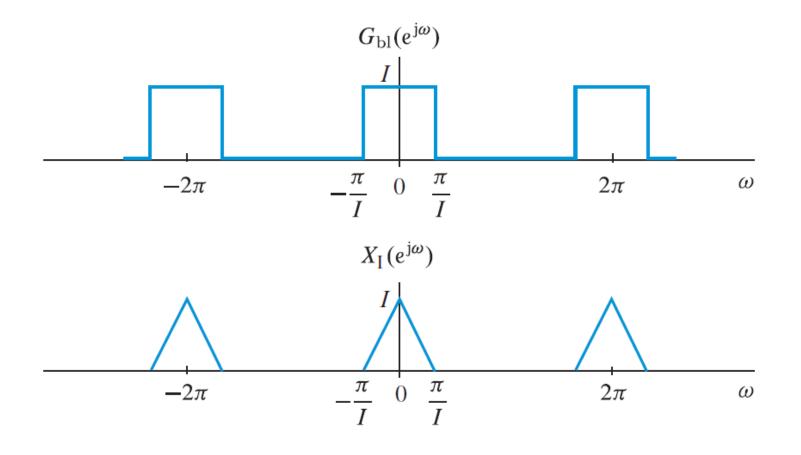
Upsample
$$x_{u}[n] \triangleq \mathcal{U}_{I}\{x[n]\} \triangleq x_{\uparrow I}x[n] \triangleq \begin{cases} x[n/I], & n \text{ is a multiple of } I \\ 0, & \text{otherwise} \end{cases}$$

DTFT $X_u(e^{j\omega}) = X(e^{j\omega l})$



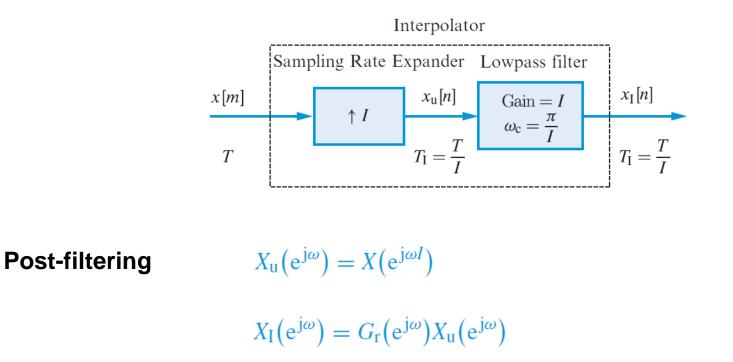


Upsampling with lowpass postfiltering (Interpolation)



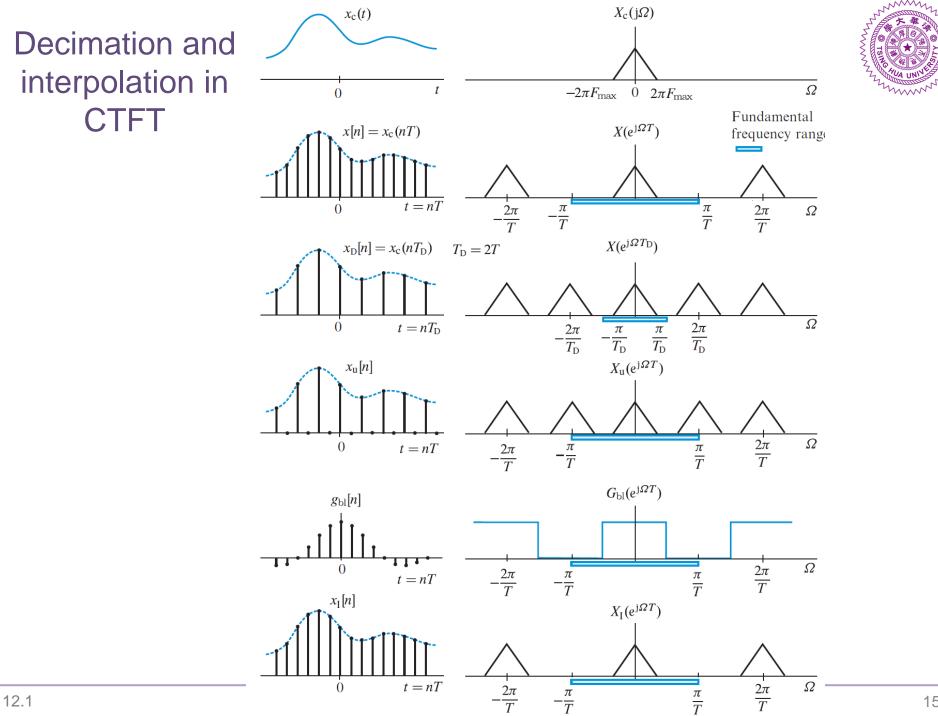
Interpolator

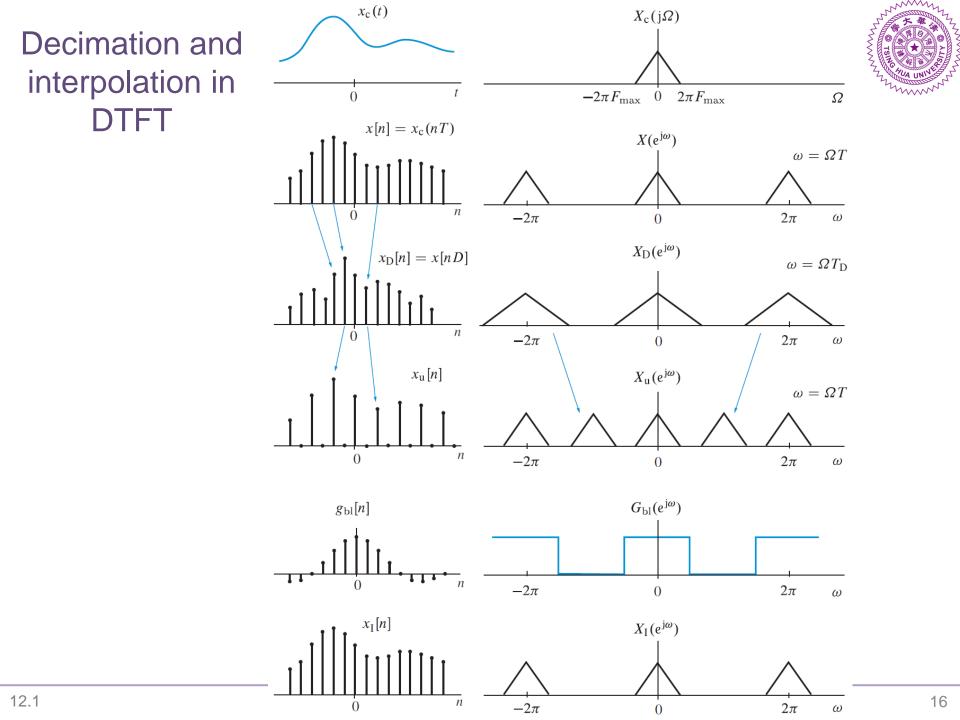




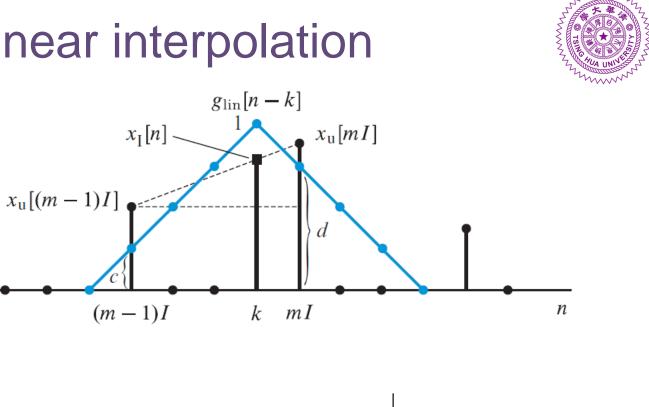
 $x_{\mathrm{I}}[n] = \sum_{k=-\infty}^{\infty} x_{\mathrm{u}}[k]g_{\mathrm{r}}[n-k]$

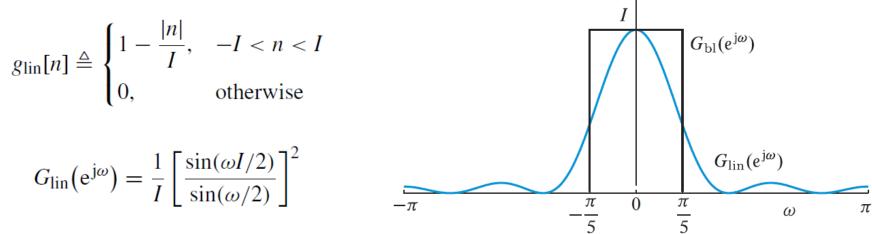
Overall computation can be reduced to 1/I for FIR filters





Linear interpolation





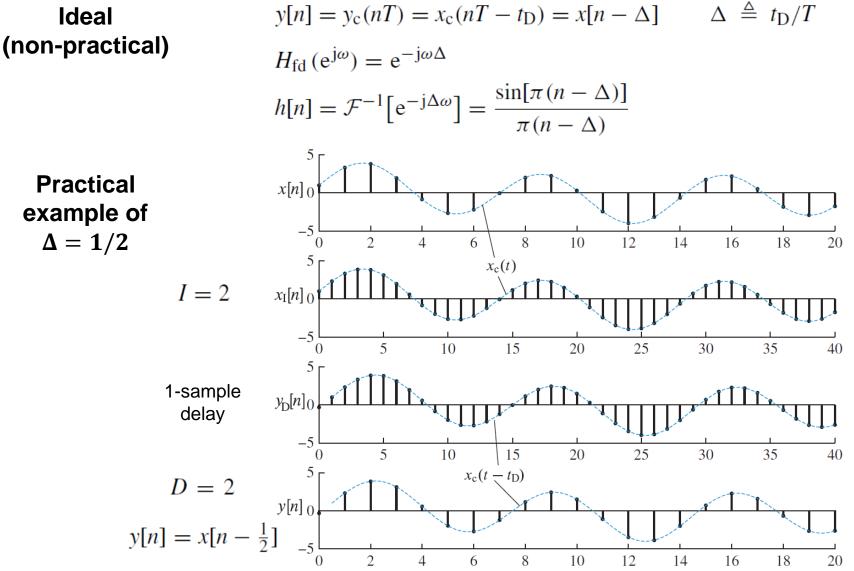
Input: Fast index

n = mI + k

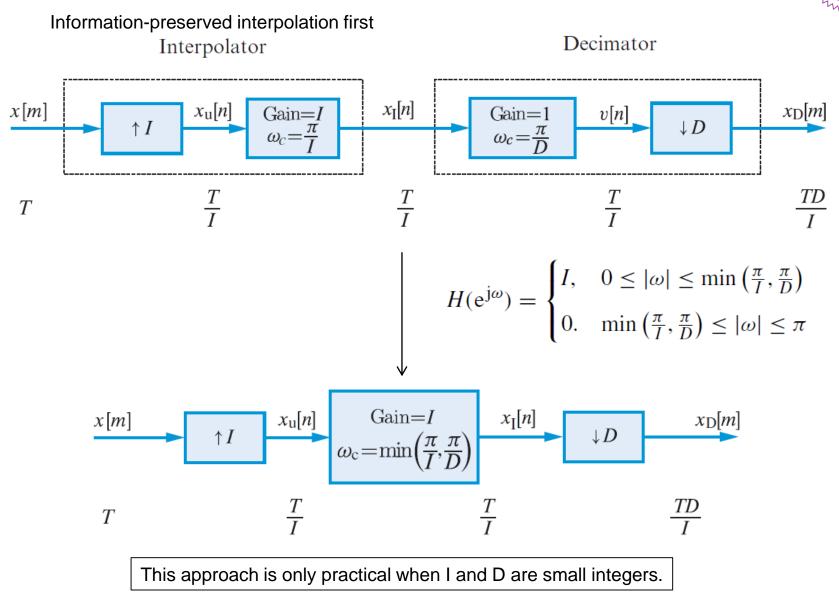
Output: Fast index

Fractional delay





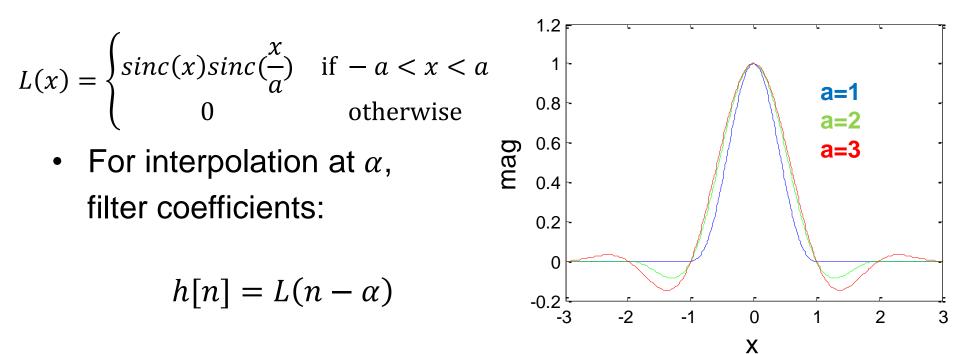
Non-integer sample rate conversion



Lanczos resampling



- Ideal sinc function masked by a sinc window
 - Good approximation for sinc function
 - Useful for interpolation, scale-up, scale-down
 - Indexed by parameter **a**



Multirate identity for downsampling



z transform of downsampling

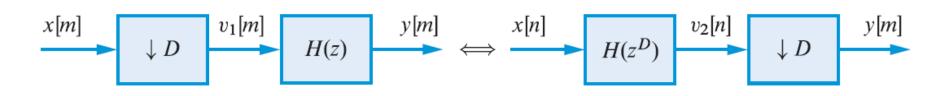
$$X_{D}(e^{j\omega}) = \frac{1}{D} \sum_{m=0}^{D-1} X\left(e^{j(\omega - 2\pi m)/D}\right)$$
$$y[n] = x[nD] \stackrel{\mathcal{Z}}{\longleftrightarrow} Y(z) = \frac{1}{D} \sum_{k=0}^{D-1} X\left(W_{D}^{k} z^{1/D}\right)$$

Interchange of filtering with downsampling

$$Y(z) = H(z)V_1(z) = H(z)\frac{1}{D}\sum_{k=0}^{D-1} X(z^{1/D}W_{\rm D}^k)$$

 $\hat{\mathbf{I}}$

$$Y(z) = \frac{1}{D} \sum_{k=0}^{D-1} V_2(z^{1/D} W_D^k) = H(z) \frac{1}{D} \sum_{k=0}^{D-1} X(z^{1/D} W_D^k) \qquad \overline{W_D^{kD}}$$



=

Multirate identity for upsampling



z transform of upsampling

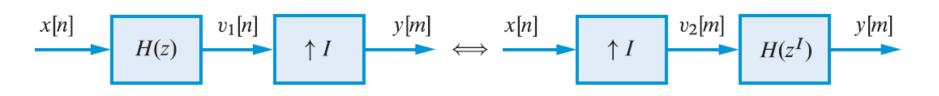
$$X_{u}(e^{j\omega}) = X(e^{j\omega I})$$
$$y[n] = \begin{cases} x[n/I], & n = 0, \pm I, \pm 2I, \dots \\ 0, & \text{otherwise} \end{cases} \longleftrightarrow Y(z) = X(z^{I})$$

Interchange of filtering with upsampling

$$Y(z) = V_1(z^I) = H(z^I)X(z^I)$$

$$\mathbf{Y}(z) = H(z^{I})V_{2}(z) = H(z^{I})X(z^{I})$$

≏



Polyphase filter structure



Example
$$H(z) = h[0] + h[1]z^{-1} + h[2]z^{-2} + h[3]z^{-3} + h[4]z^{-4} + h[5]z^{-5}$$
$$= \left(h[0] + h[2]z^{-2} + h[4]z^{-4}\right) + z^{-1}\left(h[1] + h[3]z^{-2} + h[5]z^{-4}\right)$$
$$(M=2) \qquad H(z) = P_0\left(z^2\right) + z^{-1}P_1\left(z^2\right) \qquad P_0(z) \triangleq h[0] + h[2]z^{-1} + h[4]z^{-2}$$
$$P_1(z) \triangleq h[1] + h[3]z^{-1} + h[5]z^{-2}$$
$$P_0(z) \triangleq h[0] + h[3]z^{-1}$$
$$H(z) = P_0\left(z^3\right) + z^{-1}P_1\left(z^3\right) + z^{-2}P_2\left(z^3\right) \qquad P_1(z) \triangleq h[1] + h[4]z^{-1}$$

(M=3)
$$H(z) = P_0\left(z^3\right) + z^{-1}P_1\left(z^3\right) + z^{-2}P_2\left(z^3\right) \qquad P_1(z) \triangleq h[1] + h[4]z^{-1}$$
$$P_2(z) \triangleq h[2] + h[5]z^{-1}$$

Filter Y(z) = H(z)X(z) $= P_0(z^3)X(z) + z^{-1}P_1(z^3)X(z) + z^{-2}P_2(z^3)X(z),$ $= P_0(z^3)X(z) + z^{-1}\{P_1(z^3)X(z) + z^{-1}[P_2(z^3)X(z)]\}.$



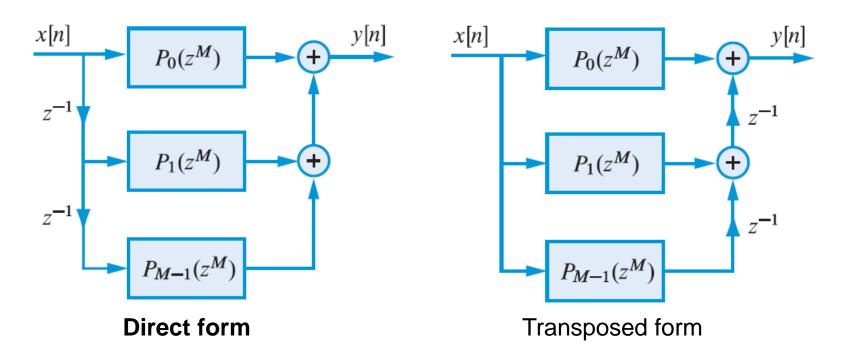
Polyphase filter structure

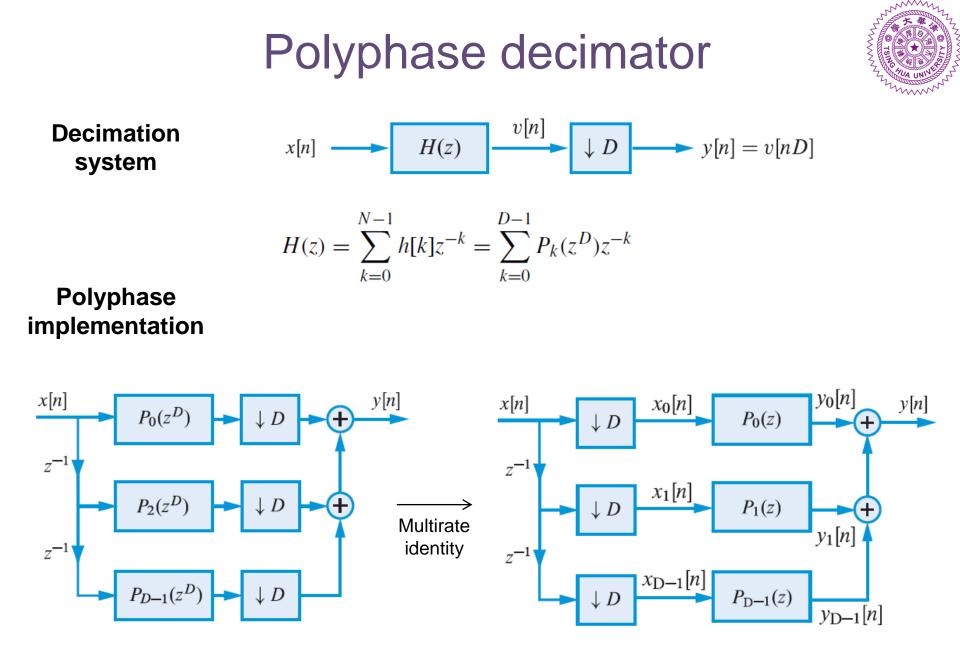
General case

$$H(z) = \sum_{k=0}^{M-1} z^{-k} P_k(z^M)$$

 $p_k[n] \triangleq h[nM+k], \quad k=0,1,\ldots,M-1.$

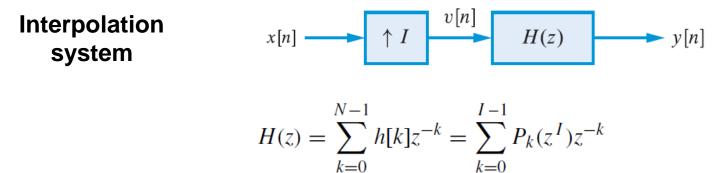
Realization



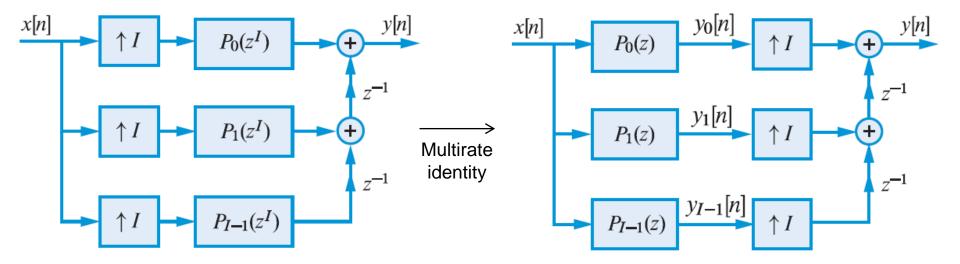


Polyphase interpolator





Polyphase implementation



Half-band filters



Ideal half-band filter (noncausal zero-phase)

$$h[n] = \frac{\omega_{\rm c}}{\pi} \left. \frac{\sin \omega_{\rm c}(n-\alpha)}{\omega_{\rm c}(n-\alpha)} \right|_{\omega_{\rm c}=\pi/2} = \begin{cases} 1/2, & n=\alpha\\ 0, & n-\alpha=\pm 2, \pm 4, \dots \end{cases}$$

General half-band filter (noncausal zero-phase)

$$h[0] = 1/2, \quad h[2n] = 0, \quad n = \pm 1, \pm 2, \dots$$

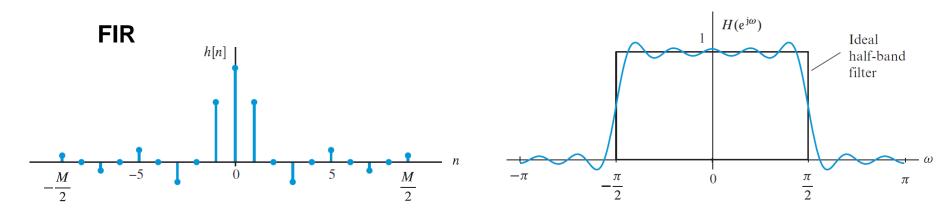
$$h[-n] = h[n] \text{ or } H(e^{-j\omega}) = H(e^{j\omega})$$

$$H(z) = P_0(z^2) + z^{-1}P_1(z^2) = \frac{1}{2} + z^{-1}P_1(z^2)$$

$$H(z) + H(-z) = 1 \qquad \qquad H(e^{j\omega}) + H(e^{j(\omega-\pi)}) = 1$$

Property

I



Half-band FIR design



1. Given the specifications ω_s , A_p , and A_s of the half-band filter, obtain the parameters δ_p , δ_s , and ω_s so that they satisfy the design requirements and the constraints of the half-band filter. That is,

$$\delta \triangleq \min(\delta_{\rm p}, \delta_{\rm s}), \quad \omega_{\rm p} = \pi - \omega_{\rm s}.$$
 (12.92)

2. Design a single band Type II FIR filter G(z) of order M/2 = 2p - 1 (odd) with $\tilde{\omega}_p = 2\omega_p$, $\tilde{\omega}_s = \pi$, and $\tilde{\delta} = 2\delta$ using the Parks-McClellan algorithm. Since G(z) is Type II, the frequency response $G(e^{j\omega})$ is equal to zero at $\omega = \pi$.

can be replaced by any filter design method, e.g. windowing

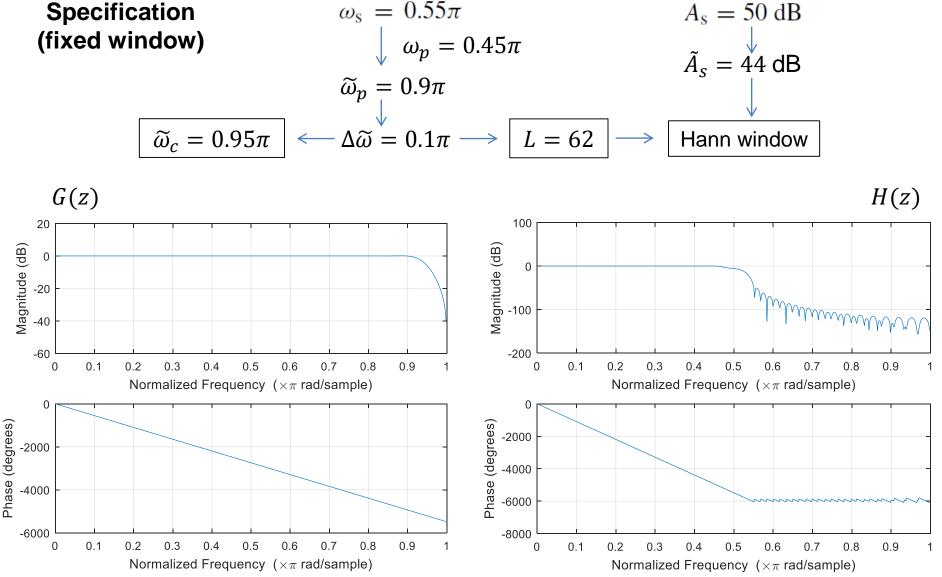
3. Scale the impulse response g[n] by one-half, upsample the result by a factor of two, and set the middle coefficient to 1/2. The result is an impulse response h[n] with system function H(z) given by

$$H(z) = \frac{1}{2} \left[z^{-M/2} + G(z^2) \right], \qquad (12.93)$$

which is a half-band filter with passband cutoff frequency ω_p .

Design example of a half-band filter





Kth-band or Nyquist filters

General Kth-band filter (noncausal zero-phase)

$$h[n] = \begin{cases} 1/K, & n = 0\\ 0, & n = \pm K, \pm 2K, \dots \end{cases}$$

$$H(z) = \frac{1}{K} + \sum_{k=1}^{K-1} z^{-k} P_k(z^K) \qquad \omega_{\rm c} = \pi/K$$

$$h[-n] = h[n]$$
 or $H(e^{-j\omega}) = H(e^{j\omega})$

Property

$$\sum_{k=0}^{K-1} H(zW_K^k) = 1 \qquad \qquad \sum_{k=0}^{K-1} H(e^{j(\omega - 2\pi k/K)}) = 1$$

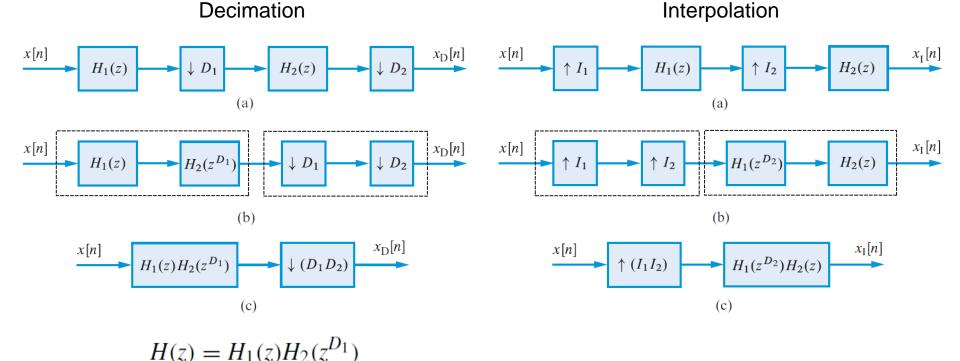
Multistage decimation and interpolation



Issue of singlestage systems Large decimation factors D (or interpolation factors I) require narrow passbands and thus demand smaller transition bands, which results in long FIR filters.

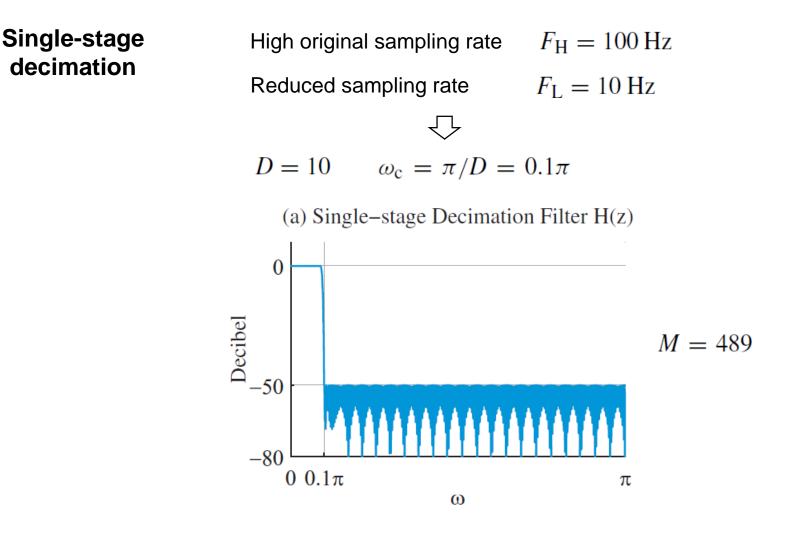
Two-stage systems

Provide shorter filters and require less computation.



Example of large decimation factor

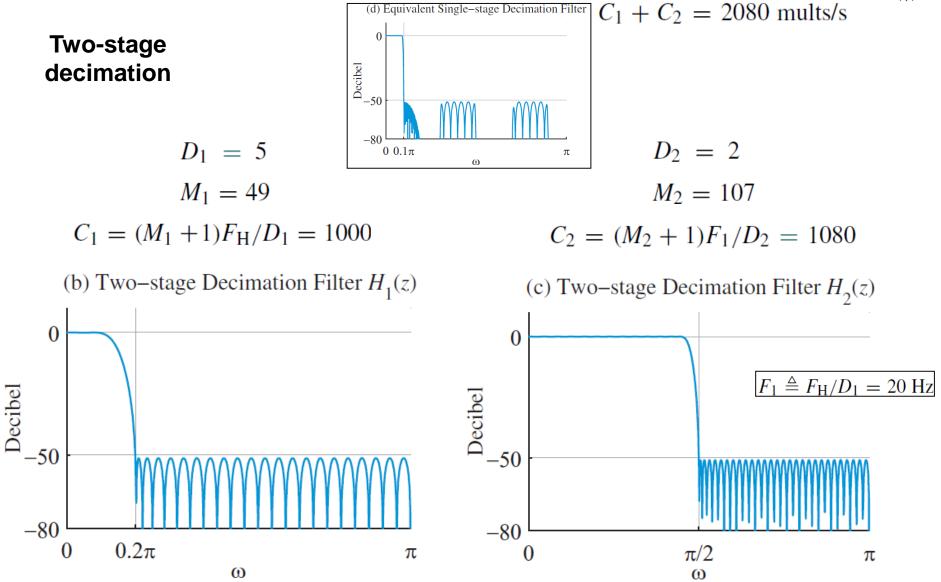


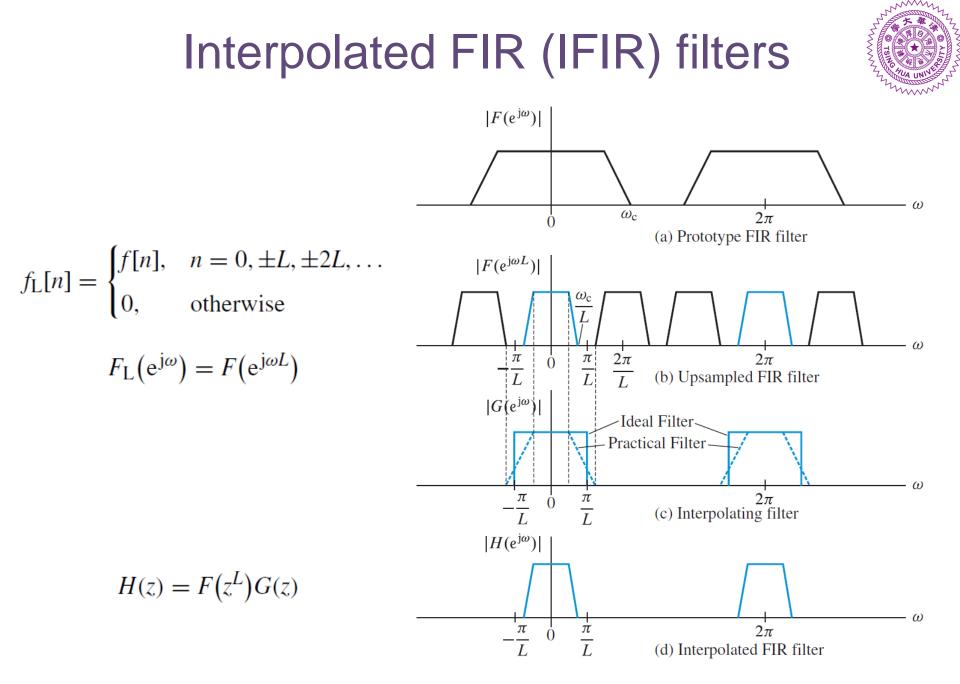


Computation complexity $C_{\rm D} = (M + 1)F_{\rm H}/D = 4900$ mults/s

Example of large decimation factor



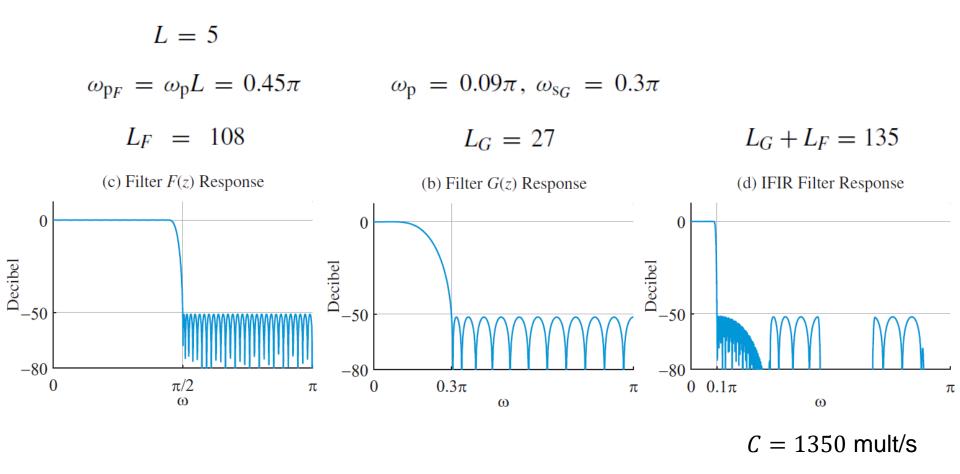




Example of large decimation factor

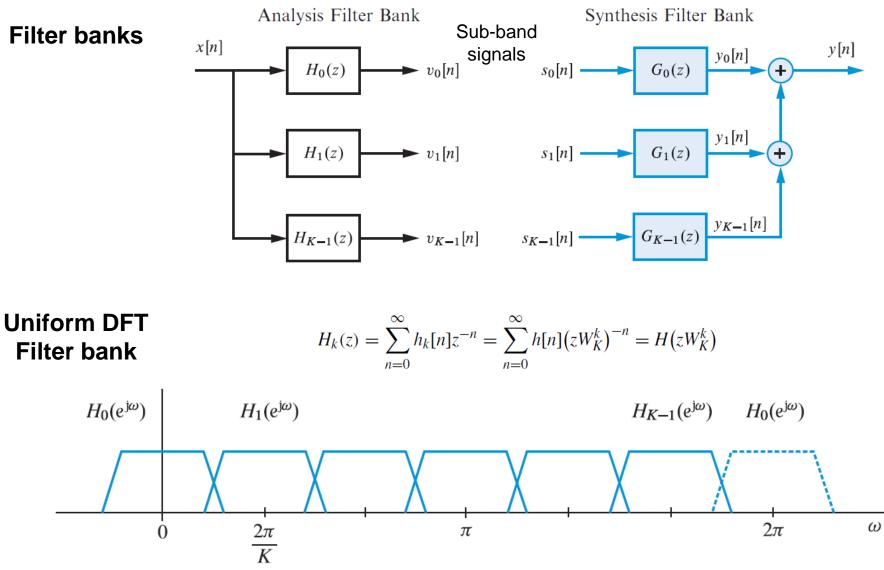


Interpolated FIR

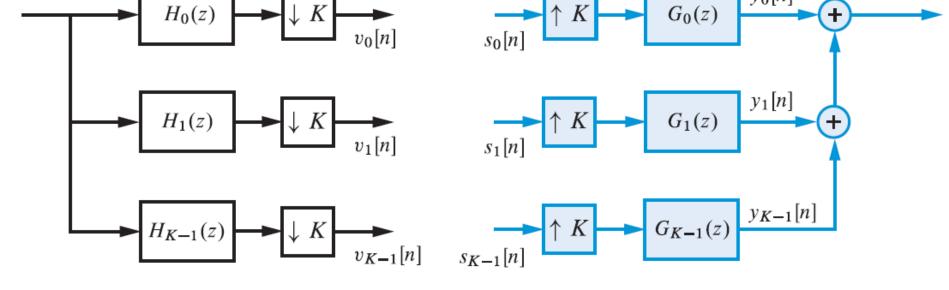


Filter bank









Analysis Filter Bank

Κ

Synthesis Filter Bank

 $y_0[n]$

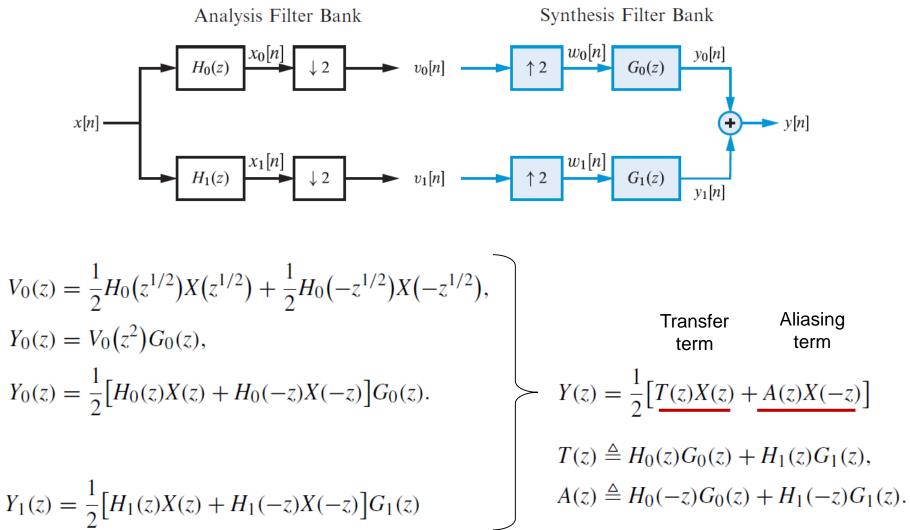


x[n]

y[n]



Two-channel filter bank





Perfect reconstruction

Condition

$$A(z) = H_0(-z)G_0(z) + H_1(-z)G_1(z) = 0,$$

$$T(z) = H_0(z)G_0(z) + H_1(z)G_1(z) = Gz^{-n_D},$$

$$\begin{bmatrix} H_0(z) & H_1(z) \\ H_0(-z) & H_1(-z) \end{bmatrix} \begin{bmatrix} G_0(z) \\ G_1(z) \end{bmatrix} = \begin{bmatrix} Gz^{-n_D} \\ 0 \end{bmatrix}$$

Solution

$$G_0(z) = \frac{2z^{-n_D}}{\Delta_m(z)} H_1(-z), \quad G_1(z) = -\frac{2z^{-n_D}}{\Delta_m(z)} H_0(-z),$$

$$\Delta_m(z) = H_0(z)H_1(-z) - H_0(-z)H_1(z) \qquad \Delta_m(z) \neq 0$$

Product filter

$$R(z) \triangleq z^{n_D} H_0(z) G_0(z) = \frac{2}{\Delta_m(z)} H_0(z) H_1(-z)$$

$$R(-z) = \frac{-2}{\Delta_m(z)} H_0(-z) H_1(z) = z^{n_D} H_1(z) G_1(z) \qquad \Delta_m(-z) = -\Delta_m(z)$$

$$R(z) + R(-z) = 2$$
 (Gain G = 2)

Two-channel perfect reconstruction filter bank



Necessary condition

$$R(z) + R(-z) = 2$$

$$\Box$$

$$R(z) = 1 + z^{-1}R_1(z^2)$$

R(z) must be a half-band filter

$$R(z) = R_0(z^2) + z^{-1}R_1(z^2)$$

$$R_0(z^2) + z^{-1}R_1(z^2) + R_0(z^2) - z^{-1}R_1(z^2) = 2$$

Proof

Additional conditions

Orthogonal

 $R(z) = H(z)H(z^{-1})$

R(z) is an autocorrelation sequence

Bi-orthogonal

 $R(z) = H_0(z)G_0(z)$

R(z) is a correlation sequence

Perfect reconstruction FIR filter bank: Conjugate quadrature filter (CQF)



CQF

 $\Delta_m(z) = H_0(z)H_1(-z) - H_0(-z)H_1(z)$ FIR condition (M is odd) $= z^{-M} \left[H(z)H(z^{-1}) + H(-z)H(-z^{-1}) \right]$ $\Rightarrow H(z)H(z^{-1}) + H(-z)H(-z^{-1}) = 1$ $|H(e^{j\omega})|^2 + |H(e^{j(\omega-\pi)})|^2 = 1$ Power complementary $|H_0(e^{j\omega})|^2 + |H_1(e^{j\omega})|^2 = 1$ $G_0(z) = 2H_1(-z) = -2z^{-M}H(z^{-1}),$ Synthesis filters $R(z) = z^{n_{\mathrm{D}}} H_0(z) G_0(z)$ $= 2H(z)H(z^{-1})$ $G_1(z) = -2H_0(-z) = -2H(-z).$

R(z) + R(-z) = 2

Properties of CQF

R(z) as autocorrelation sequence

$$r_h[n] = h[n] * h[-n] \xleftarrow{\mathcal{Z}} R_h(z) = H(z)H(z^{-1})$$
$$R_h(e^{j\omega}) = |H(e^{j\omega})|^2 \ge 0$$

Orthogonal filter

$$\sum_{k=0}^{M} h[k]h[k+2n] = 0. \quad n \neq 0 \qquad \qquad \because \text{Half-band } \mathsf{R}(\mathsf{z})$$

CQFs

$$h_0[n] = h[n] \xleftarrow{\text{DTFT}} H_0(e^{j\omega}) = H(e^{j\omega}),$$

$$h_1[n] = (-1)^n h[M - n] \xleftarrow{\text{DTFT}} H_1(e^{j\omega}) = -H(e^{-j(\omega - \pi)})e^{-j\omega M},$$

$$g_0[n] = h[M - n] \xleftarrow{\text{DTFT}} G_0(e^{j\omega}) = 2H(e^{-j\omega})e^{-j\omega M},$$

$$g_1[n] = -(-1)^n h[n] \xleftarrow{\text{DTFT}} G_1(e^{j\omega}) = -2H(e^{-j(\omega - \pi)}).$$



Design procedure of a CQF bank



- 1. Design a lowpass zero-phase half-band FIR filter $R_0(z)$ of order 2*M*, where the number *M* must be an odd integer (see Section 12.3.1).
- 2. If the minimum value δ_{\min} of the real and even function $R_0(e^{j\omega})$ is negative, form a nonnegative function as

$$R_+(e^{j\omega}) = R_0(e^{j\omega}) + |\delta_{\min}| \ge 0.$$
(12.130)

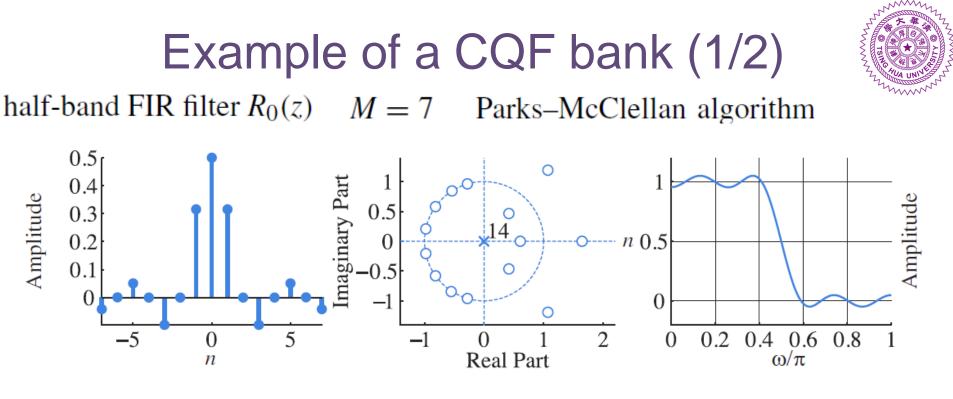
This is equivalent to adding the value $|\delta_{\min}|$ to the sample $r_0[0]$, that is,

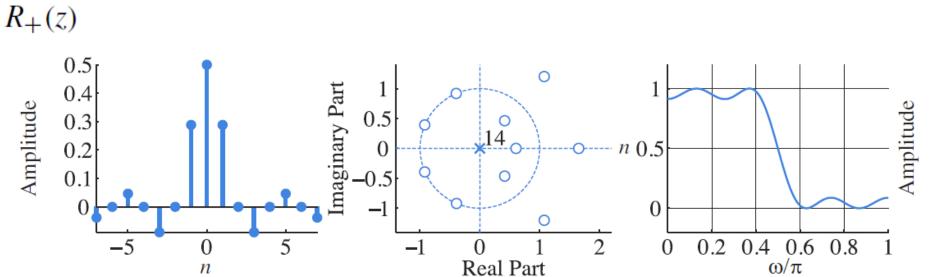
$$r_{+}[n] = r_{0}[n] + |\delta_{\min}| \,\delta[n].$$
(12.131)

3. Scale $R_+(z)$ so that the frequency response is equal to 1/2 at $\omega = \pi/2$,

$$R(z) = \frac{1/2}{1/2 + |\delta_{\min}|} R_{+}(z).$$
(12.132)

- 4. Determine the minimum-phase filter H(z) by solving the spectral factorization problem $R(z) = H(z)H(z^{-1})$ (see Section 5.8).
- 5. Specify the remaining filters of the bank using (12.115b) and (12.120).



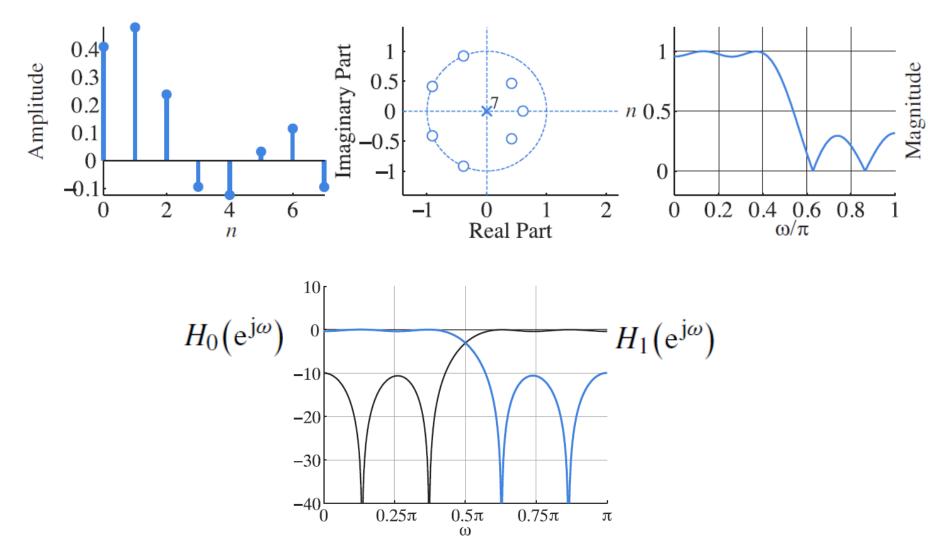


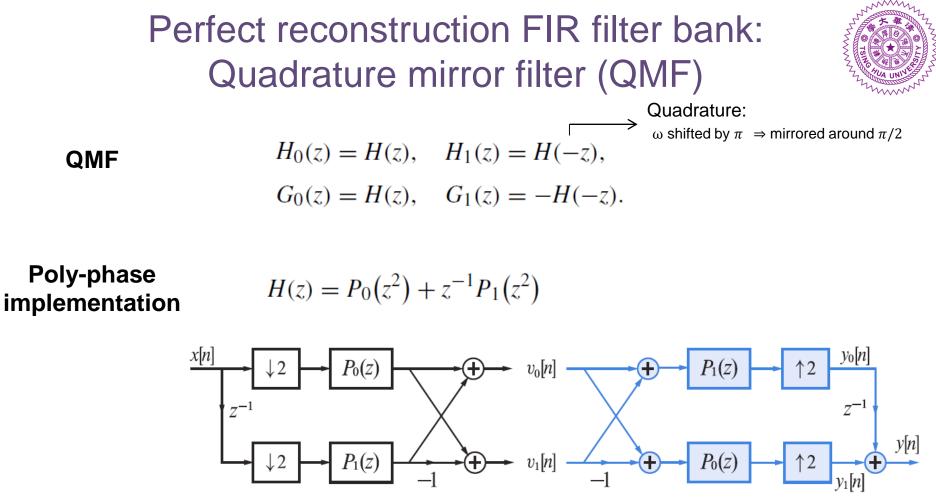
Amplitude



Example of a CQF bank (2/2)

H(z) Minimum-phase (**not linear-phase**)





Aliasing-free

 $A(z) = H_0(-z)G_0(z) + H_1(-z)G_1(z)$ = H(-z)H(z) - H(z)H(-z) = 0.

Design H(z) directly for QMF



Transfer function

 $T(z) = H_0(z)G_0(z) + H_1(z)G_1(z)$ = $H^2(z) - H^2(-z).$

$$T(z) = \Delta_m(z) = 2z^{-1}P_0(z^2)P_1(z^2)$$

PR is not practical

$$P_0(z) = b_0 z^{-n_0} \Rightarrow H(z) = b_0 z^{-2n_0} + b_1 z^{-(2n_1+1)}$$
$$P_1(z) = b_1 z^{-n_1}$$

Linear-phase H(z) (M is odd)

$$H(e^{j\omega}) = A(e^{j\omega})e^{-j\omega M/2}$$

$$\Rightarrow T(e^{j\omega}) = e^{-j\omega M} \left[\left| H(e^{j\omega}) \right|^2 + \left| H(e^{j(\omega-\pi)}) \right|^2 \right]$$

Design criterion by Johnston

$$J = \alpha \int_{\omega_{s}}^{\pi} |H(e^{j\omega})|^{2} d\omega + (1-\alpha) \int_{0}^{\pi} \left(1 - |T(e^{j\omega})|^{2}\right) d\omega$$

Example of a QMF bank 0.6 0.4 [n]0.2 0 -0.2 2 4 6 8 10 12 14 16 п Analysis filters Magnitude distortion function 0.05 r 0 Decibels Decibels -20 0 -40 -60 -0.05 0.25π 0.5π 0.75π 0.25π 0.5π 0.75π 0 0 π π ω ω