• The basic structure of the *L*-channel QMF bank is shown below



• The expressions for the *z*-transforms of various intermediate signals in the above structure are given by

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L-Channel QMF Banks

$$V_{k}(z) = H_{k}(z)X(z)$$

$$U_{k}(z) = \frac{1}{L} \sum_{\ell=0}^{L-1} H_{k}(z^{1/L}W_{L}^{\ell})X(z^{1/L}W_{L}^{\ell})$$

$$\hat{V}_{k}(z) = U_{k}(z^{L})$$

where $0 \le k \le L - 1$

• Define the vector of down-sampled subband signals as

$$\mathbf{u}(z) = \begin{bmatrix} U_0(z) & U_1(z) & \cdots & U_{L-1}(z) \end{bmatrix}^T$$

• Define the modulation vector of the input signals as

 $\mathbf{x}^{(m)}(z) = \begin{bmatrix} X(z) & X(zW_L) & \cdots & X(zW_L^{L-1}) \end{bmatrix}^T$

• Define the analysis filter bank modulation matrix as

$$\mathbf{H}^{(m)}(z) = \begin{bmatrix} H_0(z) & H_1(z) & \cdots & H_{L-1}(z) \\ H_0(zW_L) & H_1(zW_L) & \cdots & H_{L-1}(zW_L) \\ \vdots & \vdots & \ddots & \vdots \\ H_0(zW_L^{L-1}) & H_1(zW_L^{L-1}) & \cdots & H_{L-1}(zW_L^{L-1}) \end{bmatrix}$$

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- Then we can write the set of L equations $U_{k}(z) = \frac{1}{L} \sum_{\ell=0}^{L-1} H_{k}(z^{1/L} W_{L}^{\ell}) X(z^{1/L} W_{L}^{\ell})$ as $\mathbf{u}(z) = \frac{1}{L} [\mathbf{H}^{(m)}(z^{1/L})]^{T} \mathbf{x}^{(m)}(z^{1/L})$
- The output of the QMF bank is given by $Y(z) = \sum_{k=0}^{L-1} G_k(z) \hat{V}_k(z)$

• In matrix form we can write

$$Y(z) = \mathbf{g}^T(z)\mathbf{u}(z^L)$$

where

$$\mathbf{g}(z) = \begin{bmatrix} G_0(z) & G_1(z) & \cdots & G_{L-1}(z) \end{bmatrix}^T$$

• From the output equation $Y(z) = \mathbf{g}^{T}(z)\mathbf{u}(z^{L})$

the modulated versions of the output signal are given by

$$Y(zW_L^k) = \mathbf{g}^T(zW_L^k)\mathbf{u}(z^L W_L^{kL}) = \mathbf{g}^T(zW_L^k)\mathbf{u}(z^L),$$
$$0 \le k \le L - 1$$

• Define the modulation vector of the output signal as

$$\mathbf{y}^{(m)}(z) = [Y(z) \ Y(zW_L) \ \cdots \ Y(zW_L^{L-1})]^T$$

• Define the synthesis filter bank modulation matrix as

$$\mathbf{G}^{(m)}(z) = \begin{bmatrix} G_0(z) & G_1(z) & \cdots & G_{L-1}(z) \\ G_0(zW_L) & G_1(zW_L) & \cdots & G_{L-1}(zW_L) \\ \vdots & \vdots & \ddots & \vdots \\ G_0(zW_L^{L-1}) & G_1(zW_L^{L-1}) & \cdots & G_{L-1}(zW_L^{L-1}) \end{bmatrix}$$

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• Then the modulation vector of the output signal can be expressed as

$$\mathbf{y}^{(m)}(z) = \mathbf{G}^{(m)}(z)\mathbf{u}(z^L)$$

• Combining the above and $\mathbf{u}(z) = \frac{1}{L} [\mathbf{H}^{(m)}(z^{1/L})]^T \mathbf{x}^{(m)}(z^{1/L})$

we arrive at

$$\mathbf{y}^{(m)}(z) = \frac{1}{L} \mathbf{G}^{(m)}(z) [\mathbf{H}^{(m)}(z)]^T \mathbf{x}^{(m)}(z)$$

• Using the notation

$$\mathbf{T}(z) = \frac{1}{L} \mathbf{G}^{(m)}(z) [\mathbf{H}^{(m)}(z)]^T$$

we can write

$$\mathbf{y}^{(m)}(z) = \mathbf{T}(z)\mathbf{x}^{(m)}(z)$$

T(z) is called the transfer matrix relating the input signal X(z) and its modulated versions X(zW_L^k) with the output signal Y(z) and its modulated versions Y(zW_L^k)

- The filter bank is alias-free if the transfer matrix $\mathbf{T}(z)$ is a diagonal matrix of the form $\mathbf{T}(z) = \text{diag}[T(z) \ T(zW_L) \ \cdots \ T(zW_L^{L-1})]$
- The first element *T*(*z*) of the above diagonal matrix is called the distortion transfer function of the *L*-channel filter bank

• Substituting

$$V_{k}(z) = H_{k}(z)X(z)$$

$$U_{k}(z) = \frac{1}{L} \sum_{\ell=0}^{L-1} H_{k}(z^{1/L}W_{L}^{\ell})X(z^{1/L}W_{L}^{\ell})$$

$$\hat{V}_{k}(z) = U_{k}(z^{L})$$

in

$$Y(z) = \sum_{k=0}^{L-1} G_k(z) \hat{V}_k(z)$$

we arrive at

$$Y(z) = \sum_{\ell=0}^{L-1} a_{\ell}(z) X(z W_{L}^{\ell})$$

where

$$a_{\ell}(z) = \frac{1}{L} \sum_{k=0}^{L-1} H_k(zW_L^{\ell}) G_k(z), \quad 0 \le \ell \le L-1$$

• On the unit circle the term $X(zW_L^{\ell})$ becomes $X(e^{j\omega}W_L^{\ell}) = X(e^{j(\omega-2\pi\ell/L)})$

• Thus, from

$$Y(z) = \sum_{\ell=0}^{L-1} a_{\ell}(z) X(z W_{L}^{\ell})$$

we observe that the output spectrum $Y(e^{j\omega})$ is a weighted sum of $X(e^{j\omega})$ and its uniformly shifted versions $X(e^{j(\omega-2\pi\ell/L)})$ for $\ell = 1, 2, ..., L-1$ which are caused by the sampling rate alteration operations

- The term X(zW^ℓ_L) is called the ℓ-th aliasing term, with a_ℓ(z) representing its gain at the output
- In general, the *L*-channel QMF bank is a linear, time-varying system with a period *L*
- It follows from $Y(z) = \sum_{\ell=0}^{L-1} a_{\ell}(z) X(zW_{L}^{\ell})$ that the aliasing effect at the output can be completely eliminated if and only if $a_{\ell}(z) = 0, \ 1 \le \ell \le L - 1$

- Note: The aliasing cancellation condition given above must hold for all possible inputs
- If the aliasing cancellation condition holds then the *L*-channel QMF bank becomes a linear, time-invariant system with an inputoutput relation given by

$$Y(z) = T(z)X(z)$$

• The distortion transfer function T(z) is given by

$$T(z) = a_0(z) = \frac{1}{L} \sum_{k=0}^{L-1} H_k(z) G_k(z)$$

- If *T*(*z*) has a constant magnitude, then the *L*-channel QMF bank is magnitude-preserving
- If *T*(*z*) has a linear phase, then the *L*-channel QMF bank is phase-preserving
- If *T*(*z*) is a pure delay, then it is a perfect reconstruction filter bank

• Define

$$\mathbf{A}(z) = [a_0(z) \ a_1(z) \ \cdots \ a_{L-1}(z)]$$

• Then $a_{\ell}(z) = \frac{1}{L} \sum_{k=0}^{L-1} H_k(zW_L^{\ell}) G_k(z), \quad 0 \le \ell \le L-1$ can be expressed as $L \cdot \mathbf{A}(z) = \mathbf{H}^{(m)}(z) \mathbf{g}(z)$

• The aliasing cancellation condition can now be rewritten as

$$\mathbf{H}^{(m)}(z)\mathbf{g}(z) = \mathbf{t}(z)$$

where

$$\mathbf{t}(z) = \begin{bmatrix} La_0(z) & 0 & \cdots & 0 \end{bmatrix}^T$$

- Hence, knowing the set of analysis filters { $H_k(z)$ }, we can determine the desired set of synthesis filters { $G_k(z)$ } as $\mathbf{g}(z) = [\mathbf{H}^{(m)}(z)]^{-1}\mathbf{t}(z)$ provided [det $\mathbf{H}^{(m)}(z)$] $\neq 0$
- Moreover, a perfect reconstruction QMF bank results if we set $T(z) = z^{-n_o}$ in the expression for $\mathbf{t}(z)$

- In practice, the above approach is difficult to carry out for a number of reasons
- A more practical solution to the design of a perfect reconstruction QMF bank is based on a polyphase representation

- Consider the *L*-band Type I polyphase representation of the *k*-th analysis filter: $H_k(z) = \sum_{\ell=0}^{L-1} z^{-\ell} E_{k\ell}(z^L), \quad 0 \le k \le L-1$
- A matrix representation of the above set of equations is given by

$$\mathbf{h}(z) = \mathbf{E}(z^L)\mathbf{e}(z)$$

where

$$\mathbf{h}(z) = [H_0(z) \ H_1(z) \ \cdots \ H_{L-1}(z)]^T$$

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$$\mathbf{e}(z) = [1 \ z^{-1} \ \cdots \ z^{-(L-1)}]^T$$

and

$$\mathbf{E}(z) = \begin{bmatrix} E_{00}(z) & E_{01}(z) & \cdots & E_{0,L-1}(z) \\ E_{10}(z) & E_{11}(z) & \cdots & E_{1,L-1}(z) \\ \vdots & \vdots & \ddots & \vdots \\ E_{L-1,0}(z) & E_{L-1,1}(z) & \cdots & E_{L-1,L-1}(z) \end{bmatrix}$$

• **E**(*z*) is called the Type I polyphase component matrix

- Likewise, we can represent the *L* synthesis filters in a *L*-band Type II polyphase form: $G_k(z) = \sum_{\ell=0}^{L-1} z^{-(L-1-\ell)} R_{\ell k}(z^L), \quad 0 \le k \le L-1$
- In matrix form the above set of equations can be rewritten as

$$\mathbf{g}^{T}(z) = z^{-(L-1)} \tilde{\mathbf{e}}(z) \mathbf{R}(z^{L})$$

where

$$\mathbf{g}(z) = [G_0(z) \ G_1(z) \ \cdots \ G_{L-1}(z)]^T$$

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Polyphase Representation

$$\mathbf{e}(z) = \begin{bmatrix} 1 & z & \cdots & z^{L-1} \end{bmatrix} = \mathbf{e}^{T}(z^{-1})$$
and

$$\mathbf{R}(z) = \begin{bmatrix} R_{00}(z) & R_{01}(z) & \cdots & R_{0,L-1}(z) \\ R_{10}(z) & R_{11}(z) & \cdots & R_{1,L-1}(z) \\ \vdots & \vdots & \ddots & \vdots \\ R_{L-1,0}(z) & R_{L-1,1}(z) & \cdots & R_{L-1,L-1}(z) \end{bmatrix}$$

• **R**(*z*) is called the Type II polyphase component matrix

• The polyphase representations of the *L*channel analysis and the *L*-channel synthesis filter banks are shown below



Analysis filter bank

Synthesis filter bank

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• Substituting the polyphase representations of the analysis and synthesis filter banks in the original structure of the *L*-channel QMF bank, and making use of the cascade equivalences we arrive at



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• From

$$\mathbf{H}^{(m)}(z) = \begin{bmatrix} H_0(z) & H_1(z) & \cdots & H_{L-1}(z) \\ H_0(zW_L) & H_1(zW_L) & \cdots & H_{L-1}(zW_L) \\ \vdots & \vdots & \ddots & \vdots \\ H_0(zW_L^{L-1}) & H_1(zW_L^{L-1}) & \cdots & H_{L-1}(zW_L^{L-1}) \end{bmatrix}$$

and $\mathbf{h}(z) = [H_0(z) & H_1(z) & \cdots & H_{L-1}(z)]^T$
it can be seen that
 $[\mathbf{H}^{(m)}(z)]^T = [\mathbf{h}(z) & \mathbf{h}(zW_L) & \cdots & \mathbf{h}(zW_L^{L-1})]$

- Making use of $\mathbf{h}(z) = \mathbf{E}(z^L)\mathbf{e}(z)$ in the previous equation we get $[\mathbf{H}^{(m)}(z)]^T = \mathbf{E}(z^L)[\mathbf{e}(z) \ \mathbf{e}(zW_L) \ \cdots \ \mathbf{e}(zW_L^{L-1})]$
- Now, from $\mathbf{e}(z) = \begin{bmatrix} 1 & z^{-1} & \cdots & z^{-(L-1)} \end{bmatrix}^T$ we have

$$\mathbf{e}(zW_L^k) = \Delta(z) \begin{vmatrix} 1 \\ W_L^{-k} \\ \vdots \\ W_L^{-k(L-1)} \end{vmatrix}$$

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where we have used the notation $\Delta(z) = \text{diag}[1 \ z^{-1} \ \cdots \ z^{-(L-1)}]$

• Making use of the above notation in $[\mathbf{H}^{(m)}(z)]^{T} = \mathbf{E}(z^{L})[\mathbf{e}(z) \ \mathbf{e}(zW_{L}) \ \cdots \ \mathbf{e}(zW_{L}^{L-1})]$ we arrive at $\mathbf{H}(z) = \mathbf{D}^{\dagger} \Delta(z)\mathbf{E}^{T}(z^{L})$

where \mathbf{D}^{\dagger} is the conjugate transpose of the $L \times L$ DFT matrix \mathbf{D}

• Consider the *L*-channel QMF structure repeated below for convenience



• Assume that the polyphase component matrices satisfy the relation

 $\mathbf{R}(z)\mathbf{E}(z) = c\mathbf{I}$

where **I** is an $L \times L$ identity matrix and *c* is a constant

• Then the QMF structure on the previous slide reduces to the one shown on the next slide



• Note: The structure can be considered as a special case of the most general *L*-channel QMF bank shown earlier if we set $H_k(z) = z^{-k}, \ G_k(z) = z^{-(L-1-k)}, \ 0 \le k \le L-1$

• Substituting

$$H_k(z) = z^{-k}, \ G_k(z) = z^{-(L-1-k)}, \ 0 \le k \le L-1$$

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$$a_{\ell}(z) = \frac{1}{L} \sum_{k=0}^{L-1} H_k(zW_L^{\ell}) G_k(z), \quad 0 \le \ell \le L-1$$

we get $a_{\ell}(z) = z^{-(L-1)} \left(\frac{1}{L} \sum_{\ell=0}^{L-1} W_L^{-\ell k} \right), \quad 0 \le \ell \le L-1$

• Now, $\frac{1}{L}\sum_{\ell=0}^{L-1}W_L^{-\ell k} = \begin{cases} 1, & \ell = 0\\ 0, & 1 \le \ell \le L-1 \end{cases}$

• Hence, from the last equation on the previous slide it follows that

 $a_0(z) = 1, \ a_\ell(z) = 0 \ \text{for } \ell \neq 0$

• As a result, $T(z) = z^{-(L-1)}$ or in other words, the simplified QMF structure satisfies the perfect reconstruction property

- The analysis and synthesis filters of the perfect reconstruction *L*-channel QMF bank can be easily determined from known polyphase component matrices
- <u>Example</u> The structure shown below is by construction a perfect reconstruction filter bank



- The output of the filter bank is simply y[n] = dx[n-2]
- Note: In this structure $\mathbf{E}(z^3) = \mathbf{P}$ and $\mathbf{R}(z^3) = d\mathbf{P}^{-1}$
- Consider

$$\mathbf{P} = \begin{bmatrix} 1 & 1 & 1 \\ 1 & -1 & 1 \\ 1 & 0 & -1 \end{bmatrix}$$

• From $\mathbf{h}(z) = \mathbf{E}(z^3)\mathbf{e}(z)$ we get $\begin{bmatrix} H_0(z) \\ H_1(z) \\ H_2(z) \end{bmatrix} = \begin{bmatrix} E_{00}(z) & E_{01}(z) & E_{02}(z) \\ E_{10}(z) & E_{11}(z) & E_{12}(z) \\ E_{20}(z) & E_{21}(z) & E_{22}(z) \end{bmatrix} \begin{bmatrix} 1 \\ z^{-1} \\ z^{-2} \end{bmatrix}$ $= \begin{bmatrix} 1 & 1 & 1 \\ 1 & -1 & 1 \\ 1 & 0 & -1 \end{bmatrix} \begin{bmatrix} 1 \\ z^{-1} \\ z^{-2} \end{bmatrix}$

• Hence,

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$$\begin{split} H_0(z) = &1 + z^{-1} + z^{-2}, \quad H_1(z) = 1 - z^{-1} + z^{-2}, \\ &H_2(z) = 1 - z^{-2} \end{split}$$

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• With
$$d = 4$$
 we have $d\mathbf{P}^{-1} = \begin{bmatrix} 1 & 1 & 2 \\ 2 & -2 & 0 \\ 1 & 1 & -2 \end{bmatrix}$

• Then from $\mathbf{g}^{T}(z) = z^{-(L-1)} \tilde{\mathbf{e}}(z) \mathbf{R}(z^{3})$ we get $\begin{bmatrix} G_{0}(z) \\ G_{1}(z) \\ G_{2}(z) \end{bmatrix} = \begin{bmatrix} z^{-2} & z^{-1} & 1 \end{bmatrix} \begin{bmatrix} 1 & 1 & 2 \\ 2 & -2 & 0 \\ 1 & 1 & -2 \end{bmatrix}$ which leads to $G_{0}(z) = 1 + 2z^{-1} + z^{-2}, \quad G_{1}(z) = 1 - 2z^{-1} + z^{-2},$ $G_{2}(z) = -2 + 2z^{-2}$

- For a given *L*-channel analysis filter bank, the polyphase component matrix $\mathbf{E}(z)$ is known $\mathbf{E}(z)E(z) = c\mathbf{I}$,
- Hence, a perfect reconstruction *L*-channel QMF bank can be designed by constructing a synthesis filter bank with a polyphase component matrix

$$\mathbf{R}(z) = [\mathbf{E}(z)]^{-1}$$

- In general, it is not easy to compute the inverse of a rational *L*×*L* matrix
- An alternative elegant approach is to design the analysis filter bank with an invertible polyphase matrix $\mathbf{E}(z)$
- For example, $\mathbf{E}(z)$ can be chosen to be a paraunitary matrix satisfying the condition $\tilde{\mathbf{E}}(z)\mathbf{E}(z) = c\mathbf{I}$, for all z

- Note: *E*(z) is the paraconjugate of *E*(z) given by the transpose of *E*(z⁻¹), with each coefficient replaced by its conjugate
- A perfect reconstruction *L*-channel QMF bank is then obtained by choosing $\mathbf{R}(z) = \tilde{\mathbf{E}}(z)$

 For the design of a perfect reconstruction *L*-channel QMF bank, the matrix E(z) can be expressed in a product form
 E(z) = E_R(z)E_{R-1}(z)…E₁(z)E₀
 where E₀ is a constant unitary matrix, and

$$\mathbf{E}_{\ell}(z) = \mathbf{I} - \mathbf{v}_{\ell} [\mathbf{v}_{\ell}^*]^T + z^{-1} \mathbf{v}_{\ell} [\mathbf{v}_{\ell}^*]^T$$

• In the above \mathbf{v}_{ℓ} is a column vector of order *L* with unit norm, i.e., $[\mathbf{v}_{\ell}^*]^T \mathbf{v}_{\ell} = 1$

- With **E**(*z*) expressed in the product form, one can set up an appropriate objective function that can be minimized to arrive at a set of analysis filters meeting the desired specifications
- To this end, a suitable objective function is given by

$$\phi = \sum_{k=0}^{L-1} \int_{k-\text{th stopband}} \left| H(e^{j\omega}) \right|^2 d\omega$$

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- The optimization parameters are the elements of v_ℓ and E_0
- Example Consider the design of a 3channel FIR perfect reconstruction QMF bank with a passband width $\pi/3$
- The passband width of the lowpass filter is from 0 to π/3, that of the bandpass filter is from π/3 to 2π/3, and that of the highpass filter is from 2π/3 to π

• The objective function to be minimized here is thus of the form

$$\begin{split} \phi &= \int_{\frac{\pi}{3}+\varepsilon}^{\pi} \left| H_0(e^{j\omega}) \right|^2 d\omega + \int_0^{\frac{\pi}{3}-\varepsilon} \left| H_1(e^{j\omega}) \right|^2 d\omega \\ &+ \int_{\frac{2\pi}{3}+\varepsilon}^{\pi} \left| H_1(e^{j\omega}) \right|^2 d\omega + \int_0^{\frac{2\pi}{3}-\varepsilon} \left| H_2(e^{j\omega}) \right|^2 d\omega \end{split}$$

• The gain responses of the 3 analysis filters of length 15 are shown on the next slide



• The coefficients of the corresponding synthesis filters are given by $g_k[n] = h_k[14 - n], \ k = 1,2,3$