

### **SNS COLLEGE OF TECHNOLOGY An Autonomous Institution Coimbatore-35**

Accredited by NBA – AICTE and Accredited by NAAC – UGC with 'A+' Grade Approved by AICTE, New Delhi & Affiliated to Anna University, Chennai

### **DEPARTMENT OF ELECTRONICS & COMMUNICATION ENGINEERING 19ECB212 – DIGITAL SIGNAL PROCESSING**

### II YEAR/ IV SEMESTER

### **UNIT 2 – IIR FILTER DESIGN**

**TOPIC** – CHEBYSHEV FILTER

CHEBYSHEV FILTER/19ECB212 - DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT





### COMPARISON OF DIGITAL & ANALOG FILTERS

S.No.	Digital Filter	
1	Operates on digital samples of the signal	Operate
2	It is governed by linear difference equation	It is go equatio
3	It consists of adders, multipliers and delays implemented in digital logic	It considered inductor
4	The filter coefficients are designed to satisfy the desired frequency response	





**Analog Filter** 

- tes on analog signals
- overned by linear differential on
- sists of electrical components resistors, capacitors and ors
- approximation problem is satisfy the desired to ncy response



### DESIGN OF LOWPASS DIGITAL CHEBYSHEV FILTER

- For designing a Chebyshev IIR digital filter, analog filter is designed using the given specifications
- Then the analog filter transfer function is transformed to digital filter transfer  $\bullet$ function by using either Impulse Invariant or Bilinear Transformation
- The analog chebyshev filter is designed by approximating the ideal frequency response using an error function
- The approximation function is selected such that the error is minimized over a band of frequencies







### TYPES OF CHEBYSHEV & PPROXIMATION

- There are two types of Chebyshev approximation:
- 1. Type-1 Chebyshev Approximation
- 2. Type-2 Chebyshev Approximation
- In type 1 approximation, the error function is selected such that magnitude response is equiripple in the passband and monotonic in the stopband
- In type 2 approximation, the error function is selected such that magnitude response is monotonic in the passband and equiripple in the stopband
- The type 2 magnitude response is also called Inverse chebyshev response







### **ANALOG CHEBÝSHEV FILTER**

The magnitude response of Type-1 low pass filter is given by ullet

$$\left| H(j\Omega) \right|^2 = \frac{1}{1 + \epsilon^2 C_N^2 \left( \frac{\Omega}{\Omega_P} \right)}$$

*e* – Attenuation Constant

$$C_{N}\left(\frac{\Omega}{\Omega_{P}}\right) = \text{Chebyshev polynomial of or}$$

CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT





rder N



**ANALOG CHEBYSHEV FILTER** 

Attenuation Constant is given by

$$\in = \left[\frac{1}{A_P^2} - 1\right]^{\frac{1}{2}}$$

- $A_{p}$  is the gain or magnitude at pass band edge frequency  $\Omega_{p}$
- For small values of N the chebyshev polynomial is given by

$$C_{N}(\mathbf{x}) = \begin{cases} \cos(N \cos^{-1} \mathbf{x}) & ; \text{ for } |\mathbf{x}| \leq \\ \cosh(N \cosh^{-1} \mathbf{x}) & ; \text{ for } |\mathbf{x}| \end{cases}$$



# >







### **ANALOG CHEBYSHEV FILTER**

The transfer function of the analog system can be obtained from the magnitude response of Type – 1 low pass filter by substituting  $\Omega$  by s/j

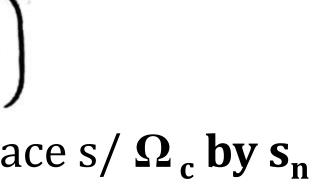
$$H(s) H(-s) = \frac{1}{1 + \epsilon^2 C_N^2 \left(\frac{s/j}{\Omega_c}\right)}$$

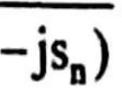
For the normalized transfer function, let us replace s/ $\Omega_{c}$  by s<sub>n</sub>

$$H(s_n) H(-s_n) = \frac{1}{1 + \epsilon^2 C_N^2}$$











### PROPERTIES OF CHEBYSHEV FILTERS (TYPE -1)

- The magnitude  $|H(j\Omega)|$  oscillates between 1 and  $1/\sqrt{1 + \epsilon^2}$  within the pass band and so the filter is called equiripple I the pass band
- The normalized magnitude response has a value of  $1/\sqrt{1 + \epsilon^2}$  at cutoff frequency  $\Omega_{c}$
- The magnitude is monotonic outside the pass band
- The Chebyshev Type 1 Filters are all pole designs
- With large values of N, the transition from pass band to stop band becomes more sharp and approaches ideal characteristics.





### TRANSFER FUNCTION OF ANALOG CHEBYSHEV LOW PASS FILTER

- For a stable and causal filter the poles should lie on the left half of s –plane. Hence the desired filter transfer function is obtained by selecting N number of left half poles
- When N is even all the poles are complex and exist as conjugate pair
- When N is odd, one of the pole is real and all other poles are complex and exist as conjugate pair
- Therefore the transfer function of chebyshev filters will be a product f second order factors





- N be the order of the filter
- H(s<sub>n</sub>) be the normalized Chebyshev lowpass filter function
- When N is even,

$$H(s_n) = \prod_{k=1}^{\frac{N}{2}} \frac{B_k}{s_n^2 + b_k s_n}$$

When N is odd,

$$H(s_n) = \frac{B_0}{s + c_0} \prod_{k=1}^{N-1} \frac{1}{s_n^2 + s_n^2}$$





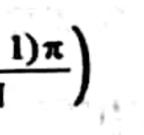


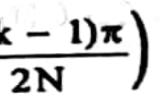
where, 
$$\mathbf{b}_{\mathbf{k}} = 2 \mathbf{y}_{\mathbf{N}} \sin\left(\frac{(2\mathbf{k} - \frac{1}{2\mathbf{N}})^2}{2\mathbf{N}}\right)$$
  
 $\mathbf{c}_{\mathbf{k}} = \mathbf{y}_{\mathbf{N}}^2 + \cos^2\left(\frac{(2\mathbf{k} - \frac{1}{2\mathbf{N}})^2}{2\mathbf{N}}\right)$   
 $\mathbf{c}_{\mathbf{0}} = \mathbf{y}_{\mathbf{N}}$   
 $\mathbf{y}_{\mathbf{N}} = \frac{1}{2} \left\{ \left[ \left(\frac{1}{\epsilon^2} + 1\right)^{\frac{1}{2}} + \frac{1}{\epsilon} \right]^{\frac{1}{2}} - \left[ \left(\frac{1}{\epsilon^2} + \frac{1}{\epsilon}\right)^{\frac{1}{2}} + \frac{1}{\epsilon} \right]^{\frac{1}{2}} \right\}$ 

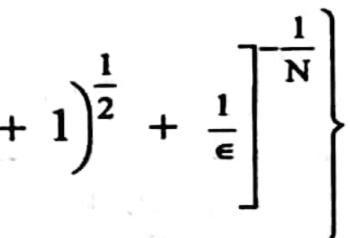
CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT













- N be the order of the filter
- For even values of N parameter  $B_k$  are evaluated

$$H(s_n)|_{s_n = 0} = \frac{1}{(1 + \epsilon^2)^2}$$

For odd values of N the parameter B<sub>k</sub> are evaluated

$$H(s_n)\Big|_{s_n=0} = 1$$

While evaluating  $B_k$  to take  $B_0 = B_1 = B_2 = \dots = B_k$ 





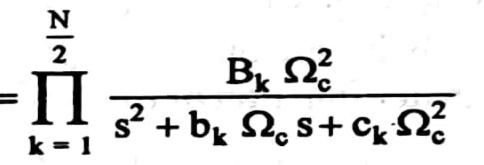




- The unnormalized transfer function is obtained by letting  $s_n \rightarrow s / \Omega_c$  in the normalized transfer function, Where  $\Omega_c$  is the cutoff frequency of the lowpass filter
- H(s) be the normalized Chebyshev low pass filter transfer function
- When N is even, H(s) is obtained by letting  $s_n \rightarrow s / \Omega_c$  in normalized Chebyshev low pass filter function

$$\therefore H(s) = \prod_{k=1}^{\frac{N}{2}} \frac{B_k}{s_n^2 + b_k s_n + c_k} \bigg|_{s_n = \frac{s}{\Omega_c}}$$



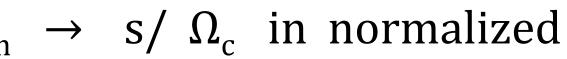


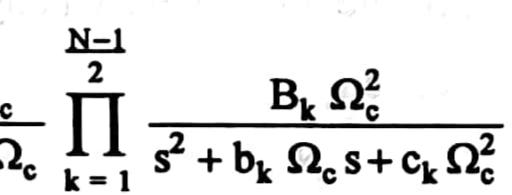


- H(s) be the normalized Chebyshev low pass filter transfer function
- N be the order of the filter
- $\Omega_c$  is the cutoff frequency of the lowpass filter
- When N is odd, H(s) is obtained by letting  $s_n \rightarrow s / \Omega_c$  in normalized Chebyshev low pass filter function

$$\therefore H(s) = \frac{B_0}{s_n + c_0} \prod_{k=1}^{\frac{N-1}{2}} \frac{B_k}{s_n^2 + b_k s_n + c_k} \bigg|_{s_n = \frac{s}{\Omega_c}} = \frac{B_0 \Omega_k}{s + c_0 \Omega_c}$$









### ORDER OF ANALOG LOWPASS CHEBYSHEV FILTER

- In Chebyshev filters the frequency response of the filter depends on the order N. The specifications of the filter are given in terms of gain at a passband and stopband frequency
- $A_{p}$  Gain or Magnitude at pass band edge frequency  $\Omega_{n}$
- $A_s$  Gain or Magnitude at Stop band edge frequency  $\Omega_s$

$$\operatorname{cosh}^{-1} \left[ \left( \frac{\left( 1/A_{s}^{2} \right) - 1}{\left( 1/A_{p}^{2} \right) - 1} \right)^{\frac{1}{2}} \right]$$
$$N_{1} = \frac{\operatorname{cosh}^{-1} \left( \frac{\Omega_{s}}{\Omega_{p}} \right)}{\operatorname{cosh}^{-1} \left( \frac{\Omega_{s}}{\Omega_{p}} \right)}$$

CHEBYSHEV FILTER/19ECB212 - DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT









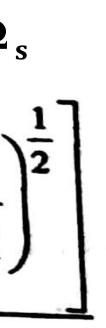
### ORDER OF THE LOWPASS CHEBYSHEV FILTER

- The specifications of the filter are given in terms of dB attenuation at a passband and stopband frequency
- $\alpha_{p, dB}$  dB attenuation at pass band edge frequency  $\Omega_{p}$
- $\alpha_{s, dB}$  dB attenuation at Stop band edge frequency  $\Omega_s$

$$\cosh^{-1}\left[\left(\frac{10^{0.1\alpha_{s,dB}}-1}{10^{0.1\alpha_{p,dB}}-1}\right)\right]$$
$$N_{1} = \frac{\cosh^{-1}\left(\frac{\Omega_{s}}{\Omega_{p}}\right)}{\cosh^{-1}\left(\frac{\Omega_{s}}{\Omega_{p}}\right)}$$

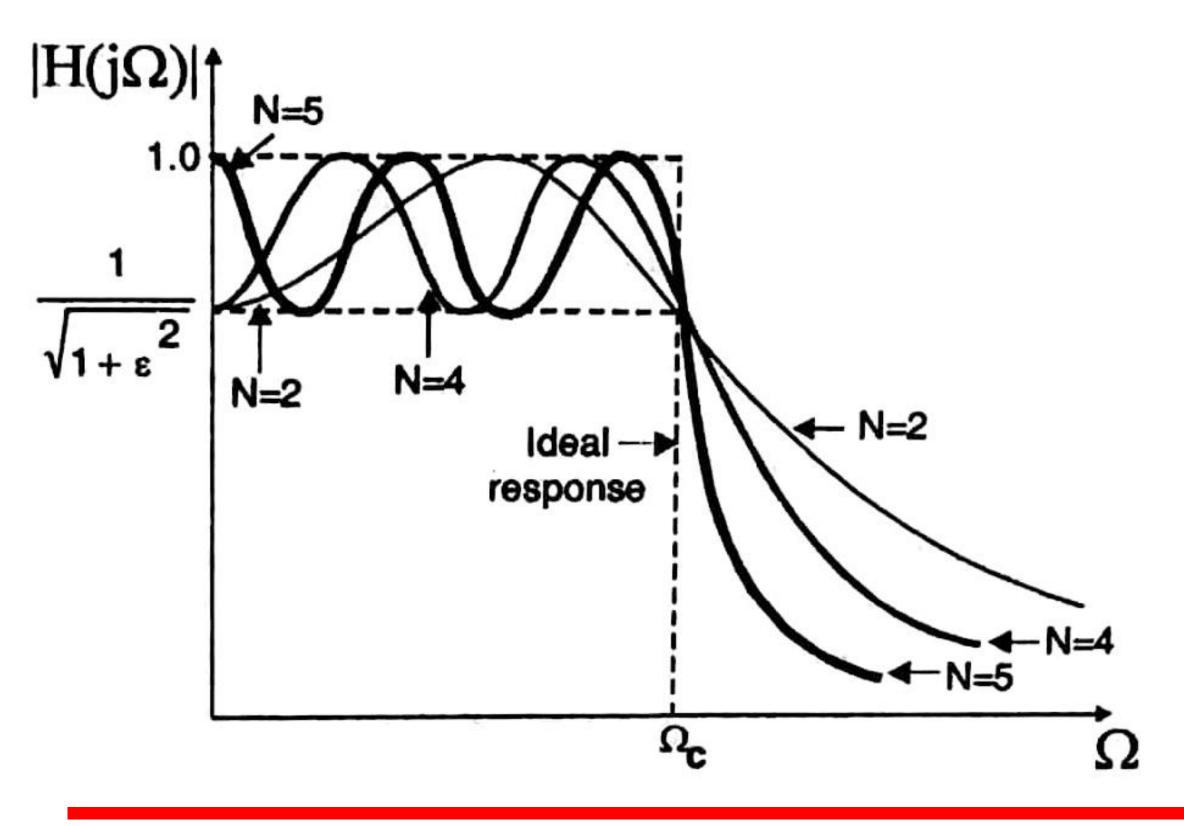








### M&GNITUDE RESPONSE OF ANALOG CHEBYSHEVTYPE – 1 LOW PASS FILTER

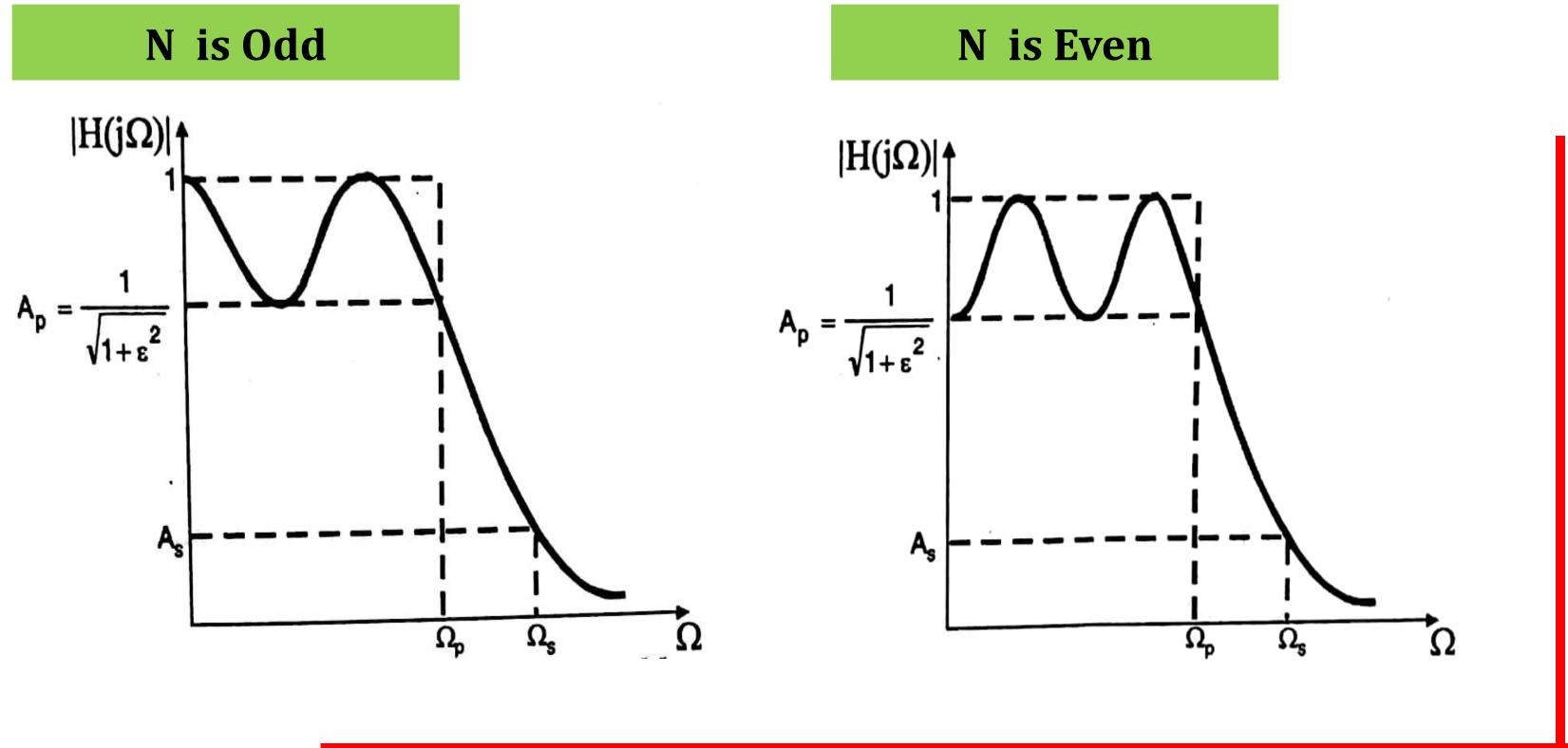


CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT



### M&GNITUDE RESPONSE OF &N&LOG CHEBYSHEV TYPE – 1 FILTERS

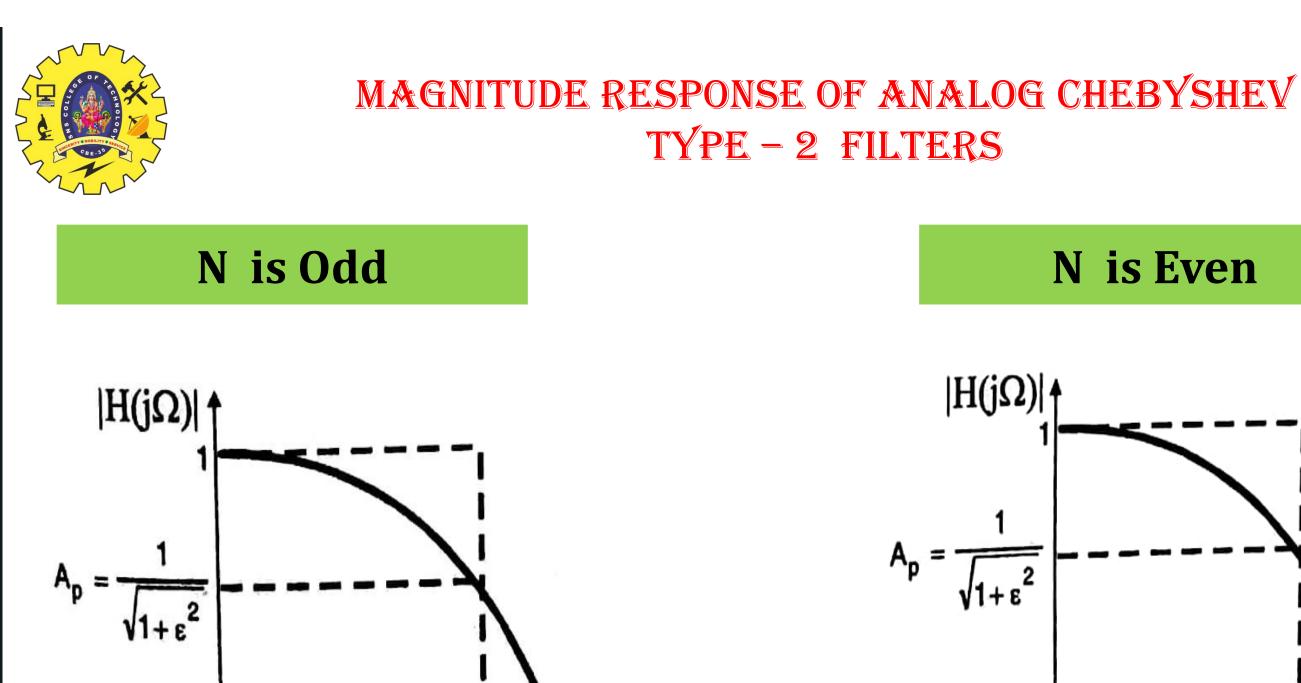


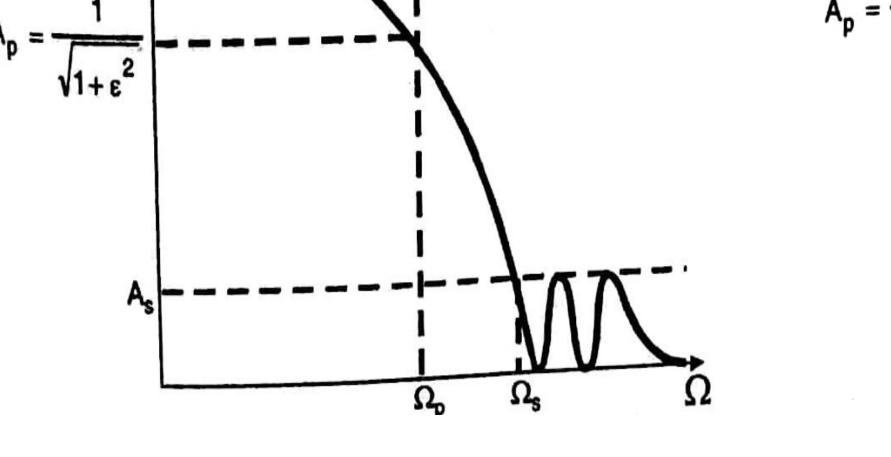


CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT



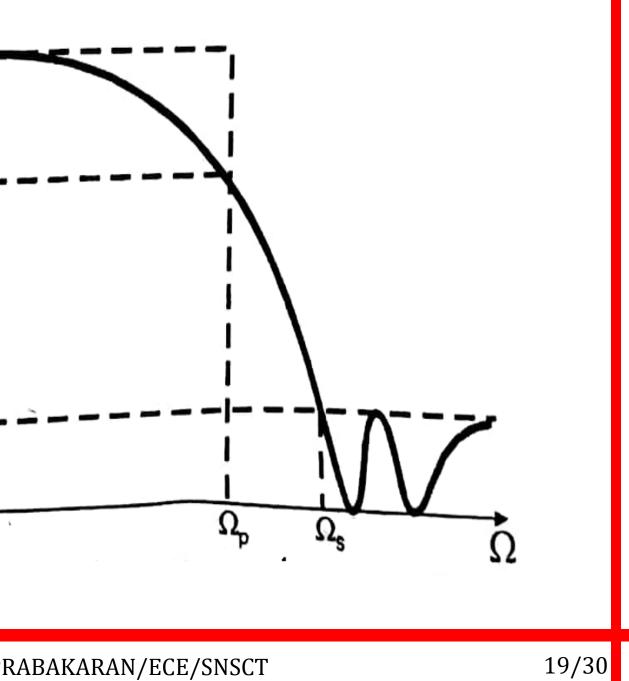






CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT







### CUTOFF FREQUENCY OF ANALOG LOWPASS CHEBYSHEV FILTER

- The IIR filters are designed to satisfy a prescribed gain or attenuation at a pass band or stop band frequency. But Practically the cutoff frequency  $\Omega_c$  is used to decide the useful frequency range of the filter
- In chebyshev filter design the passband and stopband specifications are used to estimate the order, N of the filter and N<sup>th</sup> order normalized Chebyshev lowpass filter is designed. Then the normalized LPF is unnormalized using the cutoff frequency
- In Chebyshev filters the passband edge frequency,  $\Omega_{p}$  is considered as cutoff frequency  $\Omega_c$  and this cutoff is not equal to 3 dB cutoff frequency  $\Omega_{3dB}$

$$\Omega_{3dB} = \Omega_c \cosh\left(\frac{1}{N} \cosh\right)$$





DESIGN PROCEDURE FOR LOWPASS DIGITAL CHEBYSHEV IIR FILTER

- $\omega_{p}$  Pass band edge digital frequency in rad /sample
- $\omega_s$  Stop band edge digital frequency in rad/sample
- $A_{p}$  Gain at pass band edge frequency  $\omega_{p}$
- $A_{s}$  Gain at Stop band edge frequency  $\omega_{s}$
- $T = 1/F_s$  Sampling time in sec.
- Where  $F_s =$ Sampling frequency in Hz
- $\Omega_{p}$  Pass band edge analog frequency corresponding to  $\omega_{p}$
- $\Omega_{s}$  Stop band edge analog frequency corresponding to  $\omega_{s}$





DESIGN PROCEDURE FOR LOWPASS DIGITAL CHEBYSHEV IIR FILTER

- 1. Choose either Bilinear or Impulse Invariant transformation and determine the specifications of equivalent analog filter
- The gain or attenuation of analog filter is same as digital filter
- **Bilinear Transformation:**

$$\Omega_{p} = \frac{2}{T} \tan \frac{\omega_{p}}{2} \qquad \qquad \Omega_{s} = \frac{2}{T} \tan \frac{\omega_{s}}{2}$$

**Impulse Invariant Transformation:** 

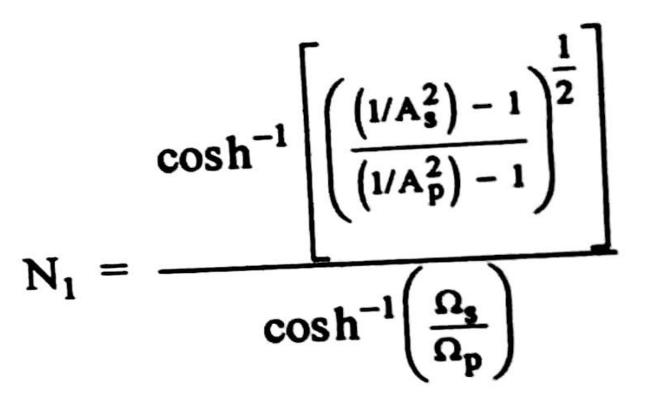
$$\Omega_{p} = \frac{\omega_{p}}{T} \qquad \Omega_{s} = \frac{\omega_{s}}{T}$$





ORDER OF THE LOWPASS DIGITAL CHEBYSHEV FILTER

2. Decide the order N of the filter. In order to estimate the order N, Calculate the Parameter  $N_1$  using the following equation:



Choose N such that,  $N \ge N_1$ . Usually N is chosen as nearest integer just greater than  $N_1$ 







NORMALIZED BUTTERWORTH LPF **TRANSFER FUNCTION** 

- 3. Determine the normalized transfer function  $H(s_n)$  of the analog lowpass filter function
- When N is even, ullet

When N is odd, ullet

$$H(s_{n}) = \prod_{k=1}^{\frac{N}{2}} \frac{B_{k}}{s_{n}^{2} + b_{k} s_{n}}$$
$$H(s_{n}) = \frac{B_{0}}{s + c_{0}} \prod_{k=1}^{\frac{N-1}{2}} \frac{\frac{N-1}{s_{n}^{2} + b_{k} s_{n}}}{\prod_{k=1}^{\frac{N-1}{2}} \frac{1}{s_{n}^{2} + b_{k} s_{n}}}$$





where, 
$$b_k = 2 y_N sin\left(\frac{(2k-1)\pi}{2N}\right)$$

$$c_k = y_N^2 + \cos^2\left(\frac{(2k-1)\pi}{2N}\right)$$

$$\mathbf{c}_0 = \mathbf{y}_N$$

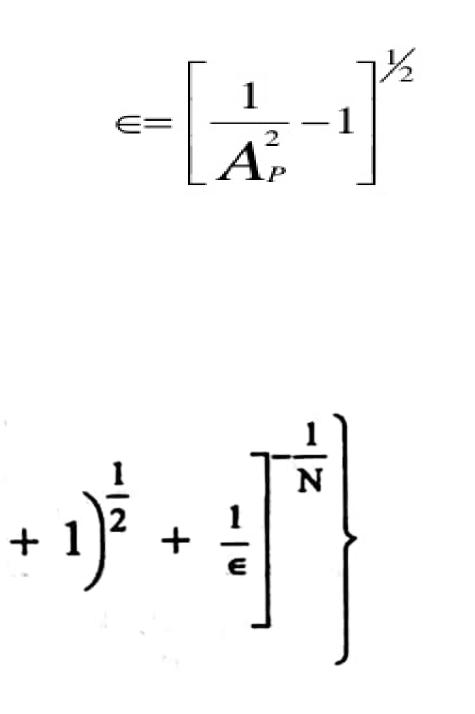
$$y_{N} = \frac{1}{2} \left\{ \left[ \left( \frac{1}{\epsilon^{2}} + 1 \right)^{\frac{1}{2}} + \frac{1}{\epsilon} \right]^{\frac{1}{N}} - \left[ \left( \frac{1}{\epsilon^{2}} + 1 \right)^{\frac{1}{2}} + \frac{1}{\epsilon} \right]^{\frac{1}{N}} \right\}$$

CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT

3-Mar-24









- N be the order of the filter
- For even values of N parameter  $B_k$  are evaluated

$$H(s_n)|_{s_n = 0} = \frac{1}{(1 + \epsilon^2)^2}$$

For odd values of N the parameter B<sub>k</sub> are evaluated

$$H(s_n)\Big|_{s_n=0} = 1$$

While evaluating  $B_k$  to take  $B_0 = B_1 = B_2 = \dots = B_k$ 









### UNNORMALIZED ANALOG TRANSFER FUNCTION

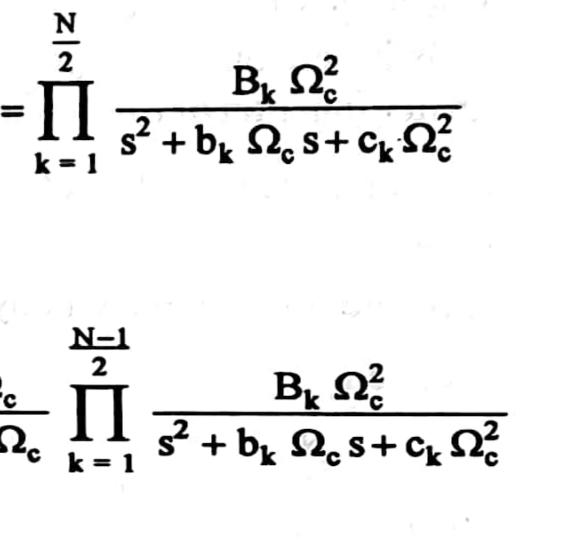
- 4. Determine the unnormalized analog transfer function H (s) is obtained by replacing  $s_n$  by s/ $\Omega_c$  in the normalized transfer function of the low pass filter function
- When N is even,

$$\therefore H(s) = \prod_{k=1}^{\frac{N}{2}} \frac{B_k}{s_n^2 + b_k s_n + c_k} \bigg|_{s_n = \frac{s}{\Omega_c}}$$

When N is odd,

$$\therefore H(s) = \frac{B_0}{s_n + c_0} \prod_{k=1}^{N-1} \frac{B_k}{s_n^2 + b_k s_n + c_k} \bigg|_{s_n = \frac{s}{\Omega_c}} = \frac{B_0 \Omega_c}{s + c_0 \Omega_c}$$







DESIGN PROCEDURE FOR LOWPASS DIGITAL CHEBYSHEV IIR FILTER

- 5. Determine the transfer function of digital filter H(z). Using the suitable transformation to transform H(s) to H(z). When the Impulse invariant transformation is employed, if T<1, then multiply H(z) by T to normalize the magnitude.
- 6. Realize the digital filter transfer function H(z) by a suitable structure
- 7. Verify the design by sketching the frequency response H ( $e^{j\omega}$ )

H (
$$e^{j\omega}$$
) = H(z) / z=  $e^{j\omega}$ 





### ASSESSMENT

- 1. What is Chebyshev approximation?
- 2. How will you choose the order N for a Chebyshev Filter?
- 3. List the Properties of Chebyshev Filter.
- 4. Compare Analog filter and Digital filter?
- 5. Analog filter transfer function is converted to a digital filter transfer function by using either ------ (or) ------
- 6. Define Sampling Time.
- 7. Attenuation Constant is given by ------
- 8. List the types of Chebyshev filters.

3-Mar-24

CHEBYSHEV FILTER/19ECB212 - DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT





## THANK YOU

3-Mar-24

CHEBYSHEV FILTER/19ECB212 – DIGITAL SIGNAL PROCESSING/J.PRABAKARAN/ECE/SNSCT



