

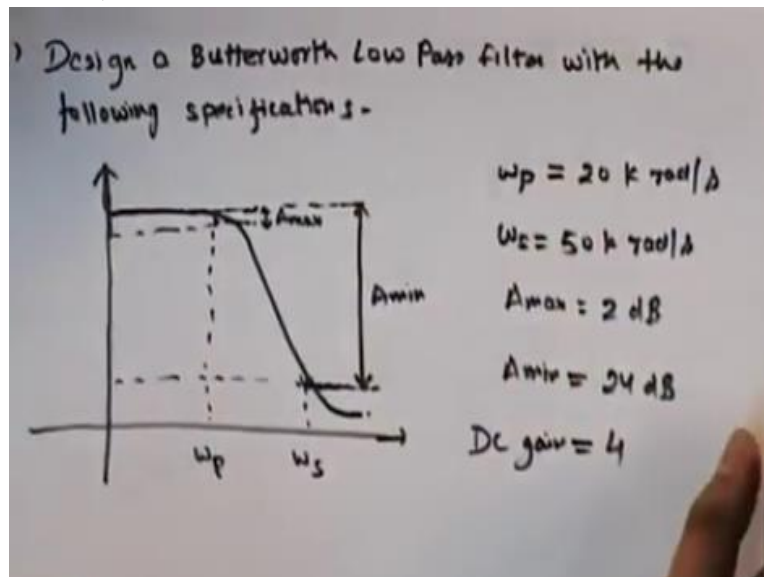
Analog Circuits
Prof. Jayanta Mukherjee
Department of Electrical Engineering
Indian Institute of Technology – Bombay

Week – 05
Module - 06
Tutorial No. 05

Hello, welcome to the next tutorial analog circuits, I am your new TA, Shashank Tiwari. In last tutorials, you have seen problems related to simple op amp circuits, phase margin, gain margin and all those things. Now in this tutorial I will be dealing with filters which are the essential components and analog circuits, so as discussed in theory sir has taught 2 kind of filters basically covering low pass filters.

These 2 filters are Butterworth low pass filter and Chebyshev low pass filter, so there is a entire design methodology for these filters, So, I will be taking 1 problem in which I will be designing the Butterworth filter of some N order that we will see what is the order of filter and then realized it and plot it just to realize that how does it come out so directly getting to the problem

(Refer Slide Time: 01:25)



The problem is design a Butterworth low pass filter with the following specifications, just to remind you I will draw the magnitude plot of the filter so I just discussed in the class there is a pass band ω_p there is a stop band ω_s this is what we call A_{max} , A_{min} is the

attenuation of DC gain at omega P. This is Amin, Amin is the DC gain attenuation at omega S, so we have been given specification in the form of omega S omega P attenuation at omega P attenuation at omega S.

So, the given omega P is omega P = 20 K radian per seconds, omega S = 50 K radian per second, Amax = 2 DB, Amin = 24 DB and the absolute DC gain we need to achieve is 4, these are all the specification given to design a filter we need to first find out the order of filter then we need to go back and find out the poles of filter.

Then after finding out the poles of filter we will be going back and seeing how to realizing in hard ware in terms of op amps circuit with RC components, I am solving this question this is the question, so I will be elaborating step by step how to solve this problem and design a filter.

(Refer Slide Time: 04:32)

$$|T(j\omega)| = \frac{K}{\sqrt{1 + \epsilon^2 \left(\frac{\omega}{\omega_p}\right)^{2N}}}$$

N = order of filter
K = dc gain

At $\omega = \omega_p$ the dc attenuation is 2 dB which means

$$20 \log \left(\frac{K}{\sqrt{1 + \epsilon^2 \left(\frac{\omega_p}{\omega_p}\right)^{2N}}} \right) = 20 \log K - 2$$

As discussed in the class there is a magnitude response of Butterworth filter which is given as $T(j\omega) = \frac{K}{\sqrt{1 + \epsilon^2 \left(\frac{\omega}{\omega_p}\right)^{2N}}}$, where N is the order of filter, K = DC gain as given in this specification that at omega P Amax is 2 DB, at omega = omega P the DC attenuation is 2 DB which means $20 \log K - 2$, so from DC gain it is $20 \log K - 2$ in DB's, going to next line.

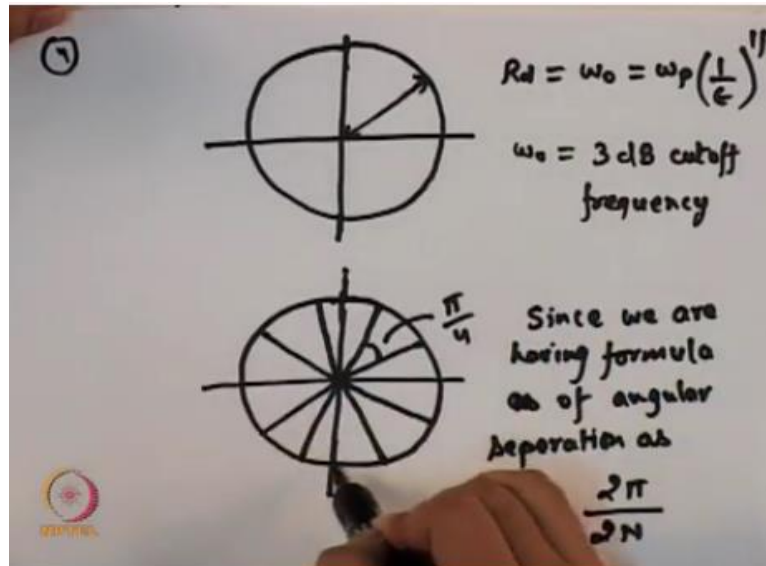
(Refer Slide Time: 01:25)

The image shows handwritten mathematical work on a whiteboard. At the top, the equation $\epsilon^2 = 10^{15} - 1$ is written. Below it, an arrow points to a boxed result $\epsilon = 0.764$. The next line is labeled "Step 2 - Finding the order of filter". Below this, the equation $20 \log \left(\frac{k}{\sqrt{1 + \epsilon^2 \left(\frac{\omega_s}{\omega_p} \right)^{2N}}} \right) = 20 \log k - 24 \text{ dB}$ is written. At the bottom, a boxed result shows $N = 3.3$.

So, after solving all this we find out that this epsilon square comes out to be 10 is to power 1 by 5 - 1 which gives epsilon = 0.764, so we have found out the epsilon so one variable we have find out, now we need to find out the order of the filter that is when, so again we have been given a specification here that at omega S = 50 K radian per second Amin which means the minimum attenuation should be 24 DB.

So, step 2: finding the order of filter, So, we have been given $20 \log K$ upon under root $1 + \epsilon^2$ at omega = omega S upon omega P raised to power $2N = 20 \log K - 24 \text{ DB}$ after calculating this and arranging the terms we will find out that the value of N would be 3.3 as we know that the order of filter cannot be fractional value it should be the integer number either it should be 3 or 4 since the minimum requirement is 3.3 which means we need to over design the filter, so I am choosing the order of filter as 4

(Refer Slide Time: 09:07)

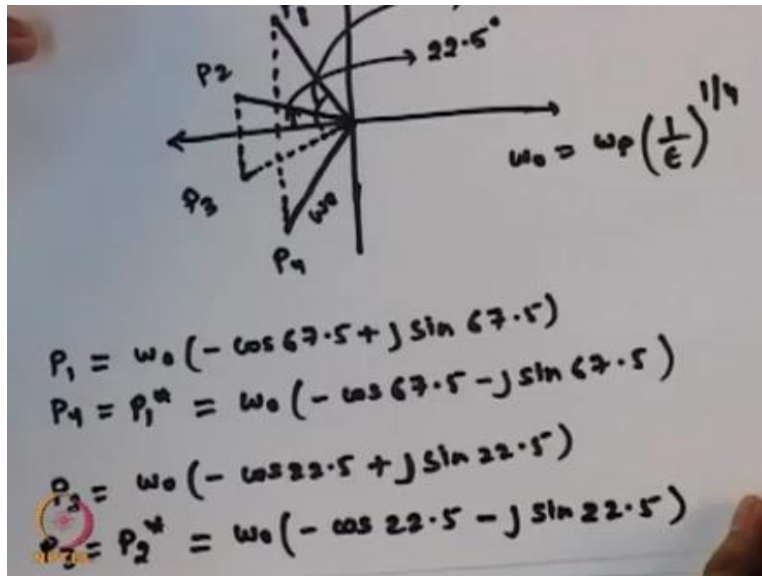


So, taking the order of filter as $N = 4$ as we know that the poles of Butterworth filter lies on a circle, so were the radius of circle $R_d = \omega_0 = \omega_p \epsilon^{-1/4}$ where 4 is the order of filter and $\omega_0 = 3 \text{ dB cutoff frequency}$, so as discussed in the class this circle will have 8 poles on it like all these poles will be equally spaced this drawing doesn't look like it is equally spaced but the angle between 2 consecutive poles is $\frac{\pi}{4}$.

Since we are having formula of angular separation as $\frac{2\pi}{2N}$, so this is the $\frac{\pi}{4}$ is the angular separation between 2 poles in this case, now let's see what are the angles we are having here, before going for that let me tell you that there are 2 kinds of poles that 4 poles are on left hand side Y-axis and 4 poles on right hand side Y-axis so for a real stable system we want always poles on the left-hand side of Y-axis.

In the discussion, I will be focusing on the poles on the left-hand side of Y-axis because the poles on the right-hand side of Y-axis belongs to the conjugate of same transfer function, so I will be dealing with the transfer function not with the conjugate of transfer function so I will be taking in to account the poles on LHS.

(Refer Slide Time: 12:20)



So, there are poles on LHS those are P1, P2, P3, P4 talking about the angular separation between 2 poles, since it is the angular separation as I told is Φ by 4 that is 45 degree, so from this negative X-axis this will be half of Φ by 4 that is 22.5 degrees and this will be 45 + 22.5 degrees this angles are from negative X axis, so from this diagram we can find out what is the P1, P2, P3, P4,

So, $P_1 = \omega_0 (-\cos 67.5 + j \sin 67.5)$ and we know it that P4 is complex conjugate of P1, so P4 would be P_1^* that is complex conjugate $\omega_0 (-\cos 67.5 - j \sin 67.5)$ just to remind that this radius is ω_0 and $\omega_0 = \omega_p \cdot \epsilon^{-1/4}$.

Theoretically we know the values of these poles, so now P2 is $\omega_0 (-\cos 22.5 + j \sin 22.5)$ and P3 is complex conjugate so it is $\omega_0 (-\cos 22.5 - j \sin 22.5)$, so these are the pole locations so half of the things are done, now the things are left using these poles we need to find out the transfer function of circuit and then realize the circuit realize the real circuit with the help of op amp and RNC's.

So, now going to next step in the class we have been briefly discussed about the second order filters, but to solve the problem further I would like to discuss this second order filter general transfer function of second order filter that is $H(s) = H_0 \frac{\omega_0^2}{s^2 + 2\zeta\omega_0 s + \omega_0^2}$.

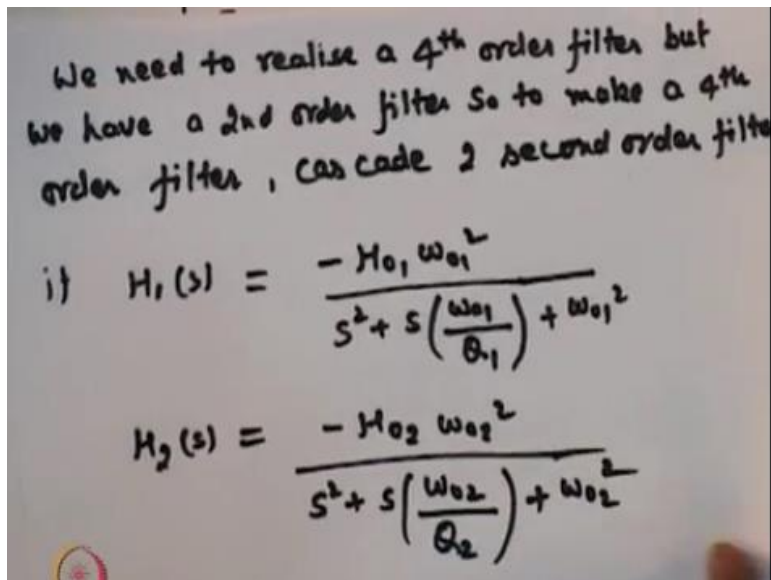
(Refer Slide Time: 16:16)

The image shows handwritten mathematical notes on a whiteboard. On the left, the transfer function is given as $H(s) = \frac{-H_0 \omega_0^2}{s^2 + \left(\frac{\omega_0}{Q}\right)s + \omega_0^2}$. Below it, the damping coefficient is defined as $\alpha = \frac{\omega_0}{2Q}$ and the imaginary part of the pole is defined as $\beta = \omega_0 \sqrt{1 - \frac{1}{4Q^2}}$. On the right, a pole-zero plot is shown in the complex s-plane. The horizontal axis is the real axis and the vertical axis is the imaginary axis. Two poles are marked as $(-\alpha, \beta)$ and $(-\alpha, -\beta)$. A dashed line connects these two poles, and its midpoint is marked with a wavy line on the real axis. The distance from the origin to the midpoint is labeled α , and the vertical distance from the real axis to either pole is labeled β . An angle ψ is shown between the real axis and the line connecting the origin to the pole $(-\alpha, \beta)$. A circled number '6' is in the top right corner.

So, this is the general transfer function of filter, now we as we know it that these poles are exist existing in complex conjugate form like P1, P4 are complex conjugate each other and P2, P4 are complex conjugate of each other, so if we are having to realize the second order filter with complex conjugate then assume this scenario that there is a pole of the filter which is $-\alpha - \beta j$ and there is a complex conjugate of this that is $-\alpha + \beta j$ and then this distance will be α and this will be β .

So, we can prove it as $\alpha = \frac{\omega_0}{2Q}$ and $\beta = \omega_0 \sqrt{1 - \frac{1}{4Q^2}}$ square this relation you can prove from this arrangement, so we have found out that the order of filter is N that is $N = 4$ and we are having a second order transfer function that is $H(s) =$ this, so how to realize of filter which is having higher order or higher order that is more than 2, so in our case will be saying that okay we are having 2 filters so will just cascade 2 filters 2 second order filters and make it make a fourth order filter.

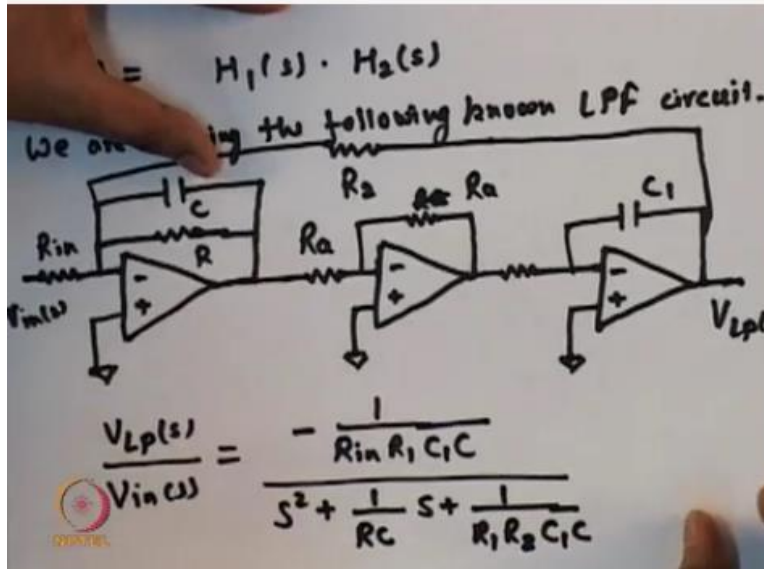
(Refer Slide Time: 18:53)



We need to realize a fourth order filter but we have a second order filter so to make a fourth order filter, cascade to second order filter suppose we are having 2 transfer functions of 2 filters like H1S and H2S so that overall transfer function of that cascaded system will be HS that is HS = H1S in to H2S.

So, we will write it like this, if H1S = - H01 omega 01 square upon S square S into omega 01 upon Q1 = omega 01 square, H2S = - 2 omega 02 square upon S Square S omega 02 upon Q2 + omega 02 square where Q1, Q2 are the quality factors of individual filters and omega 01 and omega 02 are the 3 db cutoff frequency of filter.

(Refer Slide Time: 21:28)



So, the cascaded system will have $H_S = H_1 S$ in to $H_2 S$, so this is the basic of our cascade filters so will design 2 filters 2 second order filter and cascaded to find out the overall H_S , find out the poles of system that is P_1, P_2, P_3, P_4 and then now we need to realize it, so how to realize it? What are the ways in which we can realize this system, so for that we need to take second order filter so I am taking a second order filter from text book then I will find out the component values to realize my specifications.

We are taking the following circuit following known low pass filter circuit, this is the circuit I am taking for my filter realization, so I know that transfer function of this circuit which is $V_{LPS} = V_{LPS}$ upon $V_{in} S = -1$ upon $R_{in} R_1 C_1 C$ this is R_a , upon S square 1 upon RC S upon 1 upon $R_1 R_2 C_1 C$.

(Refer Slide Time: 25:18)

Comparing this with the standard 2nd order filter equation \Rightarrow

$$\frac{-H_0 \omega_0^2}{s^2 + \left(\frac{\omega_0}{Q}\right)s + \omega_0^2}$$

We find out that

$$H_0 (\text{dc gain}) = \frac{R_2}{R_{in}} \quad \omega_0 = \frac{1}{\sqrt{R_1 R_2 C_1 C_2}}$$

$$Q = \frac{RC}{\sqrt{R_1 R_2 C_1 C_2}}$$

So, comparing this with standard second order filter equation that is $-H_0 \omega_0^2 / (s^2 + \omega_0/Q s + \omega_0^2)$ we find out that H_0 this DC gain = R_2 / R_{in} , $\omega_0 = 1 / \sqrt{R_1 R_2 C_1 C_2}$, $Q = RC / \sqrt{R_1 R_2 C_1 C_2}$.

This is the comparison with our taken circuit, so now we need to find out the DC gain from the specification we are given DC gain we will putting the values here and there and find out what is the values of $R_1 R_2 C_1 C_2$ and design our filter, so I am stopping here I will be continuing this problem in the next video.