

Control System Design
Prof. G. R. Jayanth
Department of Instrumentation and Applied Physics
Indian Institute of Science, Bangalore

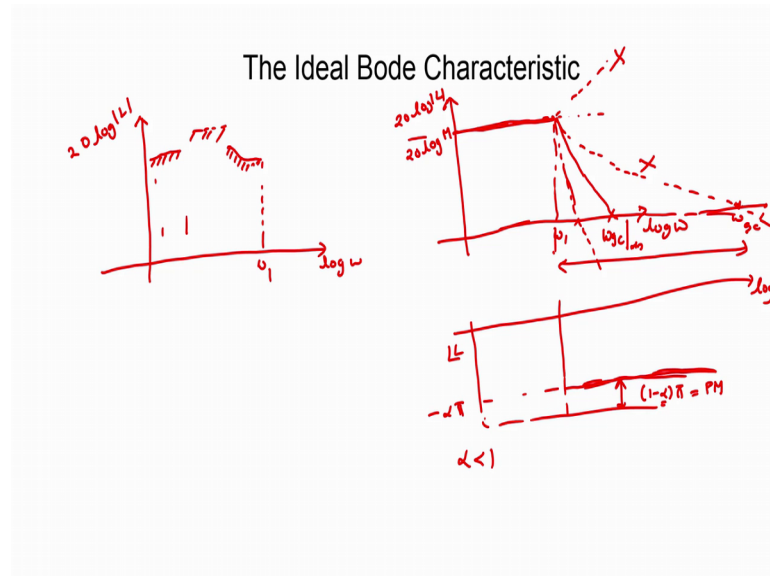
Module – 09
Lecture – 42
Ideal Bode Characteristic (Part 1/2)

Hello, the topic of today's clip is the Ideal Bode Characteristic and in this clip we shall ask ourselves an important problem that is of concern to all of us control engineers and that is what is the minimum price that we need to pay to reach the benefits of feedback. Now, what do you mean by benefits of feedback volume of certain amount of disturbance rejection you want you might want to a certain amount of robustness to variation in plant parameters you might want to track certain reference to desire level of accuracy.

So, these are all the benefit that we seek from our feedback control system and associated with this benefits we need to maintain the loop gain of our system high in the frequency ranges where this benefits are sought and what is the price that one that one needs to pay in order to realize this benefits. One has to invest in a controller which has a sufficiently high bandwidth and one has to make sure that the close to bandwidth is sufficiently high in order for us to track the references, reject disturbances, achieve robustness and so on and so forth.

So, the frequencies up to which we have to either control the shape of the loop or in other words maintain the structure of the feedback controller and the close loop bandwidth of our close loop system could count as the price of we might the paying in the for reaping the benefits that I just talked about. So, in this lecture we would try to addresses the important question as to what the minimum price we need to pay is in order to reap the benefits of feedback. So, let us try to peck this problems little bit more concrete.

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So, what I shall draw here is a Bode plot where the x axis is also omega in y axis is log of, magnitude of L $20 \log$ magnitude of L. Now, in the interest of the performance I might require $20 \log$ magnitude of L to be above I certain magnitude in one particular frequency range above a different magnitude in a different frequency range and may have to have a certain functional dependence, the magnitude need to have certain functional dependence of frequency in a third frequency range and so on and so forth.

So, this may all be part of our specification in order to either track references or reject disturbances or you know achieve robustness to plant parameter variation. Now, there will be one frequency which I shall call omega 1 at which we stop expecting any performance from our close loop system. The question is how should the loop gain appear after the frequency omega 1 so that it is most economical in terms of the price that one needs to pay to reap the benefits in the frequencies below omega 1 where we are expecting the performance specification I just talked about.

So, how should the loop shape look that is the question we are trying to answer. Now, stated as such this problem is mathematically intractable. So, you may have certain functional dependence of magnitude of loop gain as function of frequency in some frequency range, in another frequency range you might have a constant magnitude function of frequency, in a third frequency they you might have again constant magnitude, but of a different value.

So, to solve this problem of determining the shape of loop gain beyond frequency of ω_1 when you have all such complicated and competing specification of frequencies below ω_1 makes this problem difficult to tackle. So, before we solve this problem of obtaining the most economical loop gain let us first simplify the specification have been given to us and solve it for a very simple kind of specification and that itself is going to give us some important insights into the price that one needs to pay need to pay to rip the benefits of feedback.

So, rather than specify that our gain requirements or frequency dependent and they may be different and different frequencies let us assume that our gain requirement is some constant value M up to the frequency ω_1 . So, if I were to plot the Bode plot of my system then I want the gain of the open loop system to be equal to $20 \log M$ for frequencies below ω_1 , I want it to be a constant.

So, unlike the kind of dependence that the loop gain would have frequencies in the conventional case when we actually have to tackle a real life problem in order to make the analysis of the optimal loop gain mathematically tractable we shall take a step back and simplify the problem of how the loop gain is supposed to appear for frequency is below ω_1 . We shall assume it is supposed to have a magnitude again of value M in order to satisfy the various performance specifications made be disturbance rejection or reference tracking or robustness to plant parameters. Whatever it may be we shall assume that having a loop gain M will allow us to meet those specifications either exactly meet them or do better than what was expected of us.

So, we have a constant magnitude characteristic for our loop gain below frequency ω_1 and the magnitude is M and the question now that we are asking is what happened to the loop gain for frequencies beyond ω_1 , how should it look like? Well, if you think a little bit there are a few possibilities which we can easily exhaust one possibility is of course, that we allow it to increase for frequency is beyond ω_1 and clearly does naught make any sense because we are stating explicitly that there is no performance requirement beyond frequency ω_1 , so why would one want to needlessly increase the loop gain of frequency is below beyond ω_1 . There is also the price associated with controllers gain and bandwidth that needs to be paid in order to keep this loop gain is high and increased it for frequency is beyond ω_1 .

So, clearly increasing the loop gain at for frequency is beyond ω_1 makes no practical sense hence this option is altogether ruled out from our consideration. So, we have to therefore, either keep it constant which once again does not make sense because we are not expecting performance beyond frequency ω_1 and therefore, we can allow it to drop. Now, there are two ways in which loop gain can drop one is you can drop gradually and finally, crossover at some fairly high frequency and this may the gain cross over the frequency and continue to the drop over continue to decay beyond the frequency. This is one manner in which loop gain can reduce with frequency.

Now, there is one more way in which it can reduce and that is a very drastic and dramatic reduction with frequency. So, it can we can try to reduce the loop gain as quickly as possible to 0 for frequencies beyond ω_1 . Now, which of these to do we pick or should be go for some intermediate rate of decreases of loop gain with frequency. To answer this question we first shall take the case when we were allowing for fairly slow decrease in loop gain. If the rate of decreases is very low and then what happens is that our gain cross over frequencies will be quite large.

So, it will be significantly to the right in this particular graph here and what that essential means is that all the time from frequency ω_1 to frequency ω_{gc} over this entire frequency range we have to keep the loop gain high for no particular reason; we are not expected to reject any disturbances or attract any reference for frequency ω_1 .

So, needlessly we had to expand in a controller that gives this fairly large closed to band width. You need to remember the close to band width is closely related to gain cross over frequency of open loop system hence, having large gain cross over the frequency is anonymous having a large close loop bandwidth. A close loop width much larger than the frequency ω_1 up to which we are expecting performances and that does naught make any sense. There is one more reason why you do not want the gain cross over the frequency to be very large, if the gain cross over frequency is large as we just discussed the close loop band width of our system will also be exceeding the large. If our close loop band width is large then our close loop system will end up letting him lot more measurement noise than what is unavoidable.

So, in order to achieve the benefits of feedback up to frequency ω_1 , we will have to settle for feedback control and we will therefore, have to let in some measurement noise

know if we needlessly reduce the loop gain at a very slow rate then we are having a large closed loop bandwidth and over this entire closed loop bandwidth up to frequency ω_{gc} we will be letting in measurement noise. And this is want to be disastrous especially if the measurements noise has some high frequency component which will in which case I will they will be letting in significantly larger amplitudes than the case when we have a low gain crossover frequency and low closed loop bandwidth.

So, clearly this alternative of reducing the loop gain at every low rate is not a desirable one for two reasons; one is it will result in the controller having to have a fairly large bandwidth so that the closed loop system will also have a fairly large bandwidth and secondly, the fairly large bandwidth the closed loop system will make the closed loop system more sensitive to measurement noise. The output will be affected by measurement noise especially when the noise is either white noise or it has high frequency content. So, therefore, this alternative is ruled out.

How about the lost alternative that we discussed, namely where we reduced the loop gain very rapidly? In some sense this is what we desire because we are not anyway expecting performance beyond frequency ω_1 . So, the sooner we reduce the loop gain and get it to cross over the better it is. The lesser will be our close to bandwidth and therefore, the lesser will be the effect of noise on our close loop system, but can this rate of reduction of loop gain as function of frequency the arbitrarily steep, can we make this exceedingly steep?

If you think a little bit about this and go back to the previous clip where we discussed the relationship between the magnitude characteristic and a phase characteristic of a Bode plot, you might recall, but the phase characteristic of the Bode plot is related to the rate of decrease of magnitude characteristic in the Bode plot or another words the slope of the magnitude characteristic approximately gives the phase characteristics.

Now, this can tell us what is the likely price that we might pay, if we reduce the loop gain too quickly beyond the frequency ω_1 . Now, if the loop gain reduction is very fast or another word is a very steep curve then it is slop will be a very large negative number and associated with this very large reduction in slope as function of frequency you will have a very large phase lag and the phase lag by potentially big greater than minus pi radiant in which case would have phase crossing over at frequency that is less than the

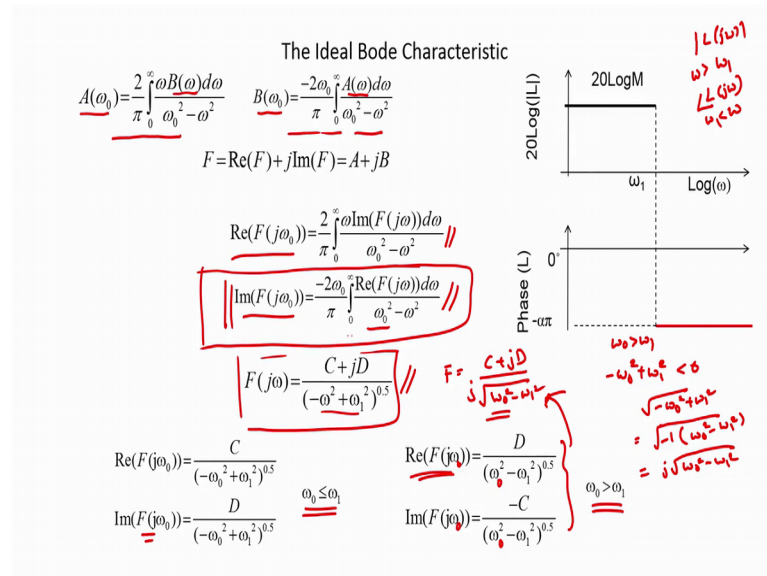
frequency at which gain cross over or in other words it would result in the close loop system by being unstable.

So, if the reduction is very steep we have this problem of excessive phase lag associated with steep reduction and that could make the system unstable if the reduction is very shallow then we end of the large closed loop system with very large bandwidth and that results in more expensive controller and more sensitivity to measurement noise. Therefore, the ideal case is to reduce it as fast as possible so, as quickly as possible, but without jeopardizing the stability of the close loop system. In other words if we assume that our phase namely angle of L should not drop below a certain value, I shall call that value as minus alpha pi where of course, alpha is less than 1, if our phase is always greater than minus alpha pi where alpha is less than 1 then we are making sure that our magnitude characteristic will at more points result in instability of the close loop system.

So, therefore, we should reduce the magnitude characteristic at the highest rate that is possible to ensure that the phase lag does not become smaller than minus alpha pi, where alpha is a number that is less than one that prevents the close loop system from becoming unstable. So, the first question that we will be looking at is to determine that minimum gain crossover frequency ω_{gc} minimum that we can have for a specified value of the frequency ω_1 and for a specified value of the magnitude M and for a specified amount of stability that we might want to have. Remember, that if minus alpha pi is the designated phase lag then the phase margin is going to be given by $1 - \alpha \pi$.

So, if I have been specified a certain phase margin or equality specified a certain alpha what is the smallest frequency at which I can get the loop gain into crossover given the frequency ω_1 up to which I am expecting performance and the magnitude M of the performance that I am expecting up to frequency ω_1 . This is the mathematical problem that we shall first end over to solve and this will give us crucial insights on the minimum price that one needs to pay in order to maintain the loop gain at M units up to frequency ω_1 . The price in a sense is the frequency range between ω_1 and ω_{gc} up to which we have to keep the loop gain high, despite the fact that there is no performance expected in this frequency range.

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So, the mathematical problem that we are trying to solve has been shown graphically on the right, we expect that the magnitude characteristic could have a constant value of 20 log M up to some frequency omega 1 and this is the frequency up to which performances is expected and beyond frequency omega 1 in the interest of stability we want the phase log one to be either greater than or equal to minus alpha pi. Now, the larger is phase log the larger is the rate of reduction of magnitude characteristic.

So, in order to get the gain cross over as early as possible beyond the frequency omega 1, we insist that the loop gain should have a phase log of exactly what is permitted, exactly the largest negative value permitted which is essentially equal to minus alpha pi. So, the question is how does the magnitude characteristic look in the frequency is beyond omega 1 that ensure that the phase log is going to be equal to minus alpha pi at frequency is beyond omega 1 and how will the phase characteristics look at frequency is below omega 1. Given that the magnitude characteristic is a constant equal to 20 log M for frequencies below omega 1 this the question that we are to trying to answer.

Once again you will be depending on some of the mathematical equipment that we have developed and used over the last couple of clips in order to determine the answer to this question of what the minimum gain crossover frequency achievable is for a specified performance and stability requirement. To answer this question as I said we shall go back to the relationship between the real and imaginary parts of a complex function F of s

which has certain special characteristics that we employed in our derivations in the previous clips.

So, for such functions we noted that the real part of the function which is $A(\omega)$ is related to the imaginary part of it by the expression given here and the imaginary part which is given by $B(\omega)$ gets determined completely if we specify $A(\omega)$ over the entire frequency range. And that relationship is given by this particular equation and I want to remind you that A and B are essentially the real and imaginary parts of a complex function F for which we carried out the derivation and obtained this result.

Now, I can rewrite the two equations here in terms of real part of F and imaginary part of F for reasons of clarity that we will get to in a minute. So, instead of writing it in terms of $A(\omega)$ and $B(\omega)$, I shall write it in terms of directly the real part of F and the imaginary part of F and that rewritten equation looks this way. Real part of $F(j\omega)$ is equal to $\frac{2}{\omega^2 - \omega_1^2}$ times integral of ω times imaginary part $F(j\omega)$ by $\omega^2 - \omega_1^2$. Imaginary part of $F(j\omega)$ is equal to minus to ω_1 times real part of $F(j\omega)$ by $\omega^2 - \omega_1^2$.

So, I shall write it in this manner and subsequently we shall define F the function F to be of the form F is equal to C plus j times D divided by $\omega^2 - \omega_1^2$ under root or divided by square root of $\omega^2 - \omega_1^2$. Now, I want to remind you before I get into further algebra and simplifications for the whole purpose of undertaking all of this algebra and integration and so on is to answer the question of how the magnitude characteristic of the loop gain. In other words how will magnitude of $L(j\omega)$ look like for frequencies ω that are greater than ω_1 and how will the angle of $L(j\omega)$ look like for frequencies ω less than the frequency ω_1 .

Having been specified that the magnitude characteristic for frequency is below ω_1 is a flat line of magnitude $20 \log M$ and the phase characteristic for frequencies is beyond ω_1 is again a flat line and to the phase lag of minus F of π . Given these magnitude and phase specifications in those particular frequency ranges how do the

magnitude and phase characteristics in a complementary ranges appear. That is the problem we are trying to solve and the mathematical equipment we were employing now is all geared to solve in this particular problem.

So, let us say we choose to apply the results of these two equations to a function F that looks something like this. F of $j\omega$ is equal to C plus jD by square root of $\omega^2 - \omega_1^2$. Now, what are these functions C of $j\omega$ and D of $j\omega$ you we will get to that in a few minutes time, for the moment though let us just assume that there are two complex functions for which these results are valid and we are now trying to apply these results for this particular function.

Now, this function is quite peculiar because the real part of F of $j\omega$ is from inspection given by C by $\sqrt{\omega^2 - \omega_1^2}$ for frequencies ω less than or equal to ω_1 . Likewise the imaginary part of F of $j\omega$ is given by D by $\sqrt{\omega^2 - \omega_1^2}$ for this in the same frequency range.

However, what is interesting is if you look at frequencies ω that are greater than ω_1 in which case we would have $\omega^2 - \omega_1^2$ to be less than 0, that happens when ω is greater than ω_1 . Then we would have that square root of $\omega^2 - \omega_1^2$ and this can be written as square root of -1 times $\sqrt{\omega^2 - \omega_1^2}$ which in essence is equal to j times square root of $\omega^2 - \omega_1^2$.

So, for frequencies ω that is greater than ω_1 we can write F as equal to C plus jD divided by j times square root of $\omega^2 - \omega_1^2$. Now, if you simplify this we will note that. So, for frequencies ω greater than ω_1 , the real part of F of $j\omega$ interestingly is going to be equal to D by $\sqrt{\omega^2 - \omega_1^2}$ and imaginary part of F of $j\omega$ is going to be equal to $-C$ by $\sqrt{\omega^2 - \omega_1^2}$. And, this is obtained directly by looking at the real and imaginary parts of the expression that has been given here.

So, we have chosen such a function F that in the frequency range up to ω_1 we have the real part of F to be given by C by $\sqrt{\omega^2 - \omega_1^2}$

root while as the frequencies greater than omega 1 we have real part of F to be given by D by omega naught square minus omega 1 square under root. And likewise also the imaginary parts the terms C and D get flipped between frequencies omega naught less than omega 1 and omega naught greater than omega 1.

The reason we are choosing this special kind of function is because we have been specified the magnitude characteristic of frequencies omega naught less than omega 1 and the phase characteristic for frequencies omega naught greater than omega 1. Suppose, we apply the second result here namely imaginary part of F of j omega naught is equal to minus 2 omega naught by pi times integral 0 to infinity real part of F of j omega D omega by omega naught square minus omega square. Suppose you apply this part to the special function F of j that we have picked then the right hand side of the expression within this integral would look something like this.

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The Ideal Bode Characteristic

$$\frac{-2\omega_0 \int_0^{\infty} \text{Re}(F(j\omega)) d\omega}{\pi \int_0^{\infty} \frac{d\omega}{\omega_0^2 - \omega^2}} = \frac{-2\omega_0 \int_0^{\omega_1} \frac{C d\omega}{(\omega_0^2 - \omega^2)(\omega_1^2 - \omega^2)^{0.5}} + \int_{\omega_1}^{\infty} \frac{D d\omega}{(\omega_0^2 - \omega^2)(\omega^2 - \omega_1^2)^{0.5}}}{\pi \int_0^{\infty} \frac{d\omega}{\omega_0^2 - \omega^2}}$$

$$\frac{-2\omega_0 \int_0^{\omega_1} \frac{C d\omega}{(\omega_0^2 - \omega^2)(\omega_1^2 - \omega^2)^{0.5}} + \int_{\omega_1}^{\infty} \frac{D d\omega}{(\omega_0^2 - \omega^2)(\omega^2 - \omega_1^2)^{0.5}}}{\pi \int_0^{\infty} \frac{d\omega}{\omega_0^2 - \omega^2}} = \begin{cases} \frac{D}{(-\omega_0^2 + \omega_1^2)^{0.5}} & \omega_0 < \omega_1 \\ \frac{C}{(\omega_0^2 - \omega_1^2)^{0.5}} & \omega_0 > \omega_1 \end{cases}$$

$$F(j\omega) = \frac{\ln L - \ln M}{(-\omega^2 + \omega_1^2)^{0.5}} = \frac{\ln |L| - \ln M + j \angle L}{(-\omega^2 + \omega_1^2)^{0.5}}$$

$$C = \ln |L| - \ln M = 0 \quad \omega_0 \leq \omega_1 \quad |L| = M \quad \omega \leq \omega_0$$

$$D = -\alpha \pi \quad \omega_0 > \omega_1$$

$$\angle L = -2\alpha \sin^{-1} \frac{\omega_0}{\omega_1} \quad \omega_0 \leq \omega_1$$

$$\ln |L| = \ln |M| - 2\alpha \ln \left[\frac{\omega_0}{\omega_1} + \sqrt{\frac{\omega_0^2}{\omega_1^2} - 1} \right] \quad \omega_0 > \omega_1$$

Handwritten notes:
 $\angle L = \angle M - 2\alpha \sin^{-1} \frac{\omega_0}{\omega_1}$
 $\ln |L| = \ln |M| - 2\alpha \ln \left[\frac{\omega_0}{\omega_1} + \sqrt{\frac{\omega_0^2}{\omega_1^2} - 1} \right]$
 $\omega_0 > \omega_1$

Minus 2 omega naught by pi integral 0 to infinity real part of F of j omega D omega by omega naught square minus omega square will look something like this because the real part of F of j omega assumes a certain functional form between the frequency 0 and omega 1 and a different functional form between the frequencies omega 1 and infinity as we just discussed in the previous slide. Indeed for frequencies up to omega 1 it is given by C by square root of omega 1 square minus omega square and for frequencies beyond omega 1 it is given by D by square root of omega square minus omega 1 square.

So, we have there for written out the right hand side of that integral as given here at this there for is going to be equal to the imaginary part of F of $j\omega$ naught and once again depending on whether ω naught is less than ω_1 or greater than ω_1 the imaginary part of F of $j\omega$ naught would be different. So, for ω naught less than ω_1 we would have the integral to be equal to D by square root of ω_1^2 minus ω naught square and for ω naught greater than ω_1 we would have the integral to be equal to minus of C by square root of ω naught square minus ω_1^2 .

Now, these are the mathematical preliminaries that we have to put in place in order to determine the magnitude characteristics of the loop gain for frequencies beyond ω_1 and for and determine the phase characteristics of frequency below ω_1 . Our goal now as control engineer is to apply it to a carefully and cannily chosen functions C and D , so that we can extract the magnitude characteristic for frequencies beyond ω_1 and phase characteristics of frequencies below ω_1 and we shall choose the function F to be of the kind F is equal to \ln of L minus \ln of M divided by square root of ω_1^2 minus ω square.

Now, since we know that \ln of L is going to be equal to \ln of magnitude of L plus j times angle of L and precisely these two terms here and we also have additional term minus \ln of M why these have this term will become evident in a few minutes time. So, from this expression we can see that the functions C of $j\omega$ is given by \ln of L minus \ln of M and interestingly we see that for frequencies ω naught less than or is equal to ω_1 this term we going to be equal to 0 because our problem specifies that the magnitude of L is equal to M for frequencies ω less than or equal to ω_1 . So, this is the part of problem specification we would have C to be equal to 0 for frequencies ω naught or less or equal to ω_1 .

Likewise for frequencies ω naught greater than ω_1 we would have the term D to be equal to minus $\alpha\pi$, where D is essentially equal to the angle of L that is going to equal to minus $\alpha\pi$ of frequencies ω naught greater than ω_1 or more generally for frequencies ω greater than ω_1 . So, we can employ these two realizations in simplifying the appearance of this particular integral. We note that the first integral here goes from 0 to ω_1 and it integrates the term C by square root of

$\omega_1^2 - \omega^2$ times $\omega_0^2 - \omega^2$ and denote that C is equal to 0 within the limits 0 to ω_1 and hence what we would naught is that this integral essentially goes to 0.

As far as the second integral is concerned we would have that the term D would be a constant for frequencies between ω_1 and infinity and this term D essentially would be equal to $-\alpha\pi$ in this frequency range because that is the phase lag that we have assumed for our loop gain between ω_1 and infinity in order to make sure that in our attempt to reduce our loop gain as quickly as possible. We are not going to be destabilizing the closed loop system.

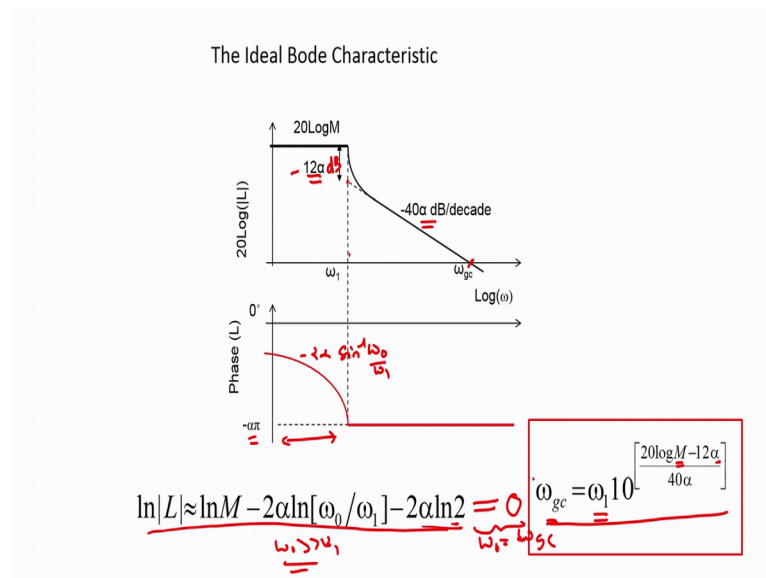
So, if we substitute that, then the integral on the left hand side gets reduced to expression that has been shown here and on the right side for frequencies ω_0 less than ω_1 we would have it to be equal to D by square root of $\omega_1^2 - \omega_0^2$. And noting that for the particular F of $j\omega$ that we have to chosen the function D is essentially given by angle of L , it is going to be equal to angle of L by square root of $\omega_1^2 - \omega_0^2$ for frequencies is ω_0 less than or equal to ω_1 and it is going to be equal to minus of \ln of magnitude of L minus \ln of M divided by square root of $\omega_0^2 - \omega_1^2$ for frequencies ω_0 greater than ω_1 . Because it is going to be equal to essentially minus of C by square root of $\omega_0^2 - \omega_1^2$ according to this expression and see our particular problem is essentially \ln of magnitude of L minus \ln of M and that is how you get this particular expression.

Now, we can solve this integral to directly obtained the angle of L for frequencies ω_0 less than or is equal to ω_1 and the magnitude of the L for frequencies ω_0 greater than ω_1 . Now, the procedure to solve this particular integrally relatively straight forward and it is given in the reference that I have indicated in the notes and if one word compute this integral namely 2α times ω_1 to infinity D by $\omega_0^2 - \omega_1^2$ times square root of $\omega_1^2 - \omega_0^2$. We would note that the angle of L upon solving the integral is going to be given by $-\alpha \sin^{-1}(\omega_0/\omega_1)$ and \ln of magnitude of L is going to given by \ln of M minus $2\alpha \ln$ of ω_0/ω_1 plus square

root of omega naught square by omega 1 square minus 1 for frequencies omega naught greater than omega 1.

So, we have succeeding obtaining the angle of L as function of frequency, for frequencies omega naught less than or is equal to omega 1 and the magnitude of L as function of frequencies for frequencies beyond omega 1. So, if you going to graph the angle of L the magnitude of L in these respective frequencies ranges it will look something like this.

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In the frequencies range beyond omega 1 we would have magnitude characteristic rolling of and the role of rate is going to given by minus 40 alpha D be per decade where alpha pi was the specified phase lag of our open loop system and phase response for frequencies below omega 1 is given by minus 2 alpha sin inverse omega naught by omega 1. Now, to check whether the phase response is right, we note that if you substitute omega naught equal to omega 1 we get that the phase lag at exactly the frequency omega 1 is going to be equal to minus alpha pi. And, therefore, there is going to be no discontinuity in our phase estimate for frequencies is beyond omega 1 and exactly at omega 1.

So, if we return to the magnitude characteristic we note that this magnitude characteristic causes over at a certain frequency and this frequency is a smallest frequency at which we can allow the magnitude characteristic to crossover. That is because if the magnitude

characteristic were to cross over any frequency less than this frequency then the associated the phase lag could be so large that the requirement that the phase lag be at most minus alpha pi would be violated.

So, how do we obtain the gain cross over the frequency ω_{gc} ? In order to obtain ω_{gc} let us first make an approximation to the magnitude characteristic of frequencies beyond ω_1 . If we written to the expression of $\ln L$ we have that \ln of magnitude of L is equal to \ln of M minus 2α times ω_0 by ω_1 plus square root of ω_0 square by ω_1 square minus 1. Now, in the frequency range ω_0 greater than ω_1 , we would have that ω_0 by ω_1 will be much better than 1 and hence the term within the bracket here can be simplified us \ln of magnitude of L is equal to \ln of M minus 2α times \ln of 2 times ω_0 by ω_1 .

And, I have written the same approximate expression here. So, for frequencies ω_0 which is much better than ω_1 we would have \ln of magnitude of L to be approximately equal to \ln of M minus $2\alpha \ln$ of ω_0 by ω_1 minus $2\alpha \ln 2$. And when we convert the x axis to the as log of frequency to the base 10 and the y axis as log of magnitude of L to the base 10 times 20 the term $2\alpha \ln 2$ essentially becomes equal to 12α decibels. And, the role of is given by minus 2α if the x axis and y axis for simply \ln of magnitude of L and \ln of frequency, but if we were to inspect plot $20 \log$ to the base 10 magnitude of L verses \log to the 10 base of frequency the slope which earlier minus to α essentially becomes minus 40α dB per decade.

So, the magnitude characteristic roles of approximately linearly for frequency is ω_0 that is much greater than ω_1 and if were to extend this linear characteristic up to the frequency ω_1 we would notice that linear trend will intersect the vertical line at ω_1 at a point that is 12α dB lower than the $20 \log M$. Hence in order to determine the gain crossover frequency we need to note that at the gain crossover frequency you would have $20 \log$ of magnitude of L to be equal to be 0 or equivalently \ln of magnitude of L will also be equal to 0. So, we just have to set this entire expression to be equal to 0 to obtain the frequency ω_{gc} .

So, what ω_{gc} would equal to ω_1 you would have this to be valid, namely this entire \ln of magnitude of L will be equal to 0. Now, if we want to solve this equation we get the minimum gain crossover frequency that is given by this expression ω_{gc} minimum is equal to $\omega_1 10^{\frac{20 \log M - 12\alpha}{40}}$. So, if we specified a certain phase margin requirement then we fix the value for α and if we specify the performance requirements in terms of the extent of disturbance rejection or the maximum permissible error in tracking we essentially specify the term M . And with these two pieces of information along with the information of the frequency ω_1 up to which the performance is expected we can compute the minimum gain crossover frequency that is possible for our close loop system for it to achieve the performance M up to frequency ω_1 and achieve a phase margin as characterized by the constant α .

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$\frac{\pi}{4} = (1-\alpha)\pi$ $-\frac{5\pi}{4}$

- Example:** Assuming $(M=100)$ and $PM=30^\circ$, it is seen that $\alpha = 5/6$ and that $\omega_{gc} = 8\omega_1$. For $PM=45^\circ$, i.e., $\alpha = 3/4$, $\omega_{gc} = 11\omega_1$.

$PM = \frac{\pi}{4} = (1-\alpha)\pi$

So, to drive home the significance of this fact let us take a numerical example. So, let us assume that as control engineers we are expected to achieve a gain of 100 which corresponds to 40 dB and up to some frequency ω_1 and have a phase margin of at least 30 degrees. Now, we note that if the phase margin is 30 degrees or in the other words $5/6$ radian, then we would have that the maximum phase lag which is characterized by the term $-\alpha\pi$ can be obtained by solving the equations $\pi/6$ is equal to $1 - \alpha$ times π and if rearrange it we get that α is equal to $5/6$ in

other words the maximum log permitted in order for us to achieve a phase margin of 30 degrees will be minus $5\pi/6$.

So, for this particular permitted phase lag and for this particular magnitude requirement up to some frequency ω_1 , we obtain the gain cross over the frequency the minimum gain cross over the frequency to be 8 times ω_1 . So, almost nearly order of magnitude greater than ω_1 . Now, suppose we want a slightly more conservative design and desire a phase margin of 45 degrees or in other words the phase margin is going to equal to $\pi/4$, then we can show that α can be calculated as $1 - \alpha$ times $\pi/4$ is equal to the phase margin or is equal to $\pi/4$ and that gives α to be equal to $3/4$.

Now, we substitute this particular value for α along with the particular required value for the gain M in the equation that I showed in the previous slide you will obtain the gain crossover frequency to be equal to 11 times ω_1 . So, what this indicates to us in general is that as a rule of thumb the price that we pay for achieving the benefits of feedback up to a frequency ω_1 is that we have to keep the loop gain high, in other words greater than 0 dB for a frequency range that is nearly 10 times larger than the frequency ω_1 up to which we are expecting performance and this is the best that we can do.

Any smaller value of gain cross over frequency and hence the closed loop bandwidth would result in a phase lag that is going to be lesser than minus $\alpha\pi$ in the phase margin requirement is will be violated. So, as a rule of thumb therefore, the frequency range over which we pay price or in other words the closed loop bandwidth of our system of our control system is in general going to be about 10 times larger than the frequency range up to which were expecting performance from our close loop system.

Thank you.