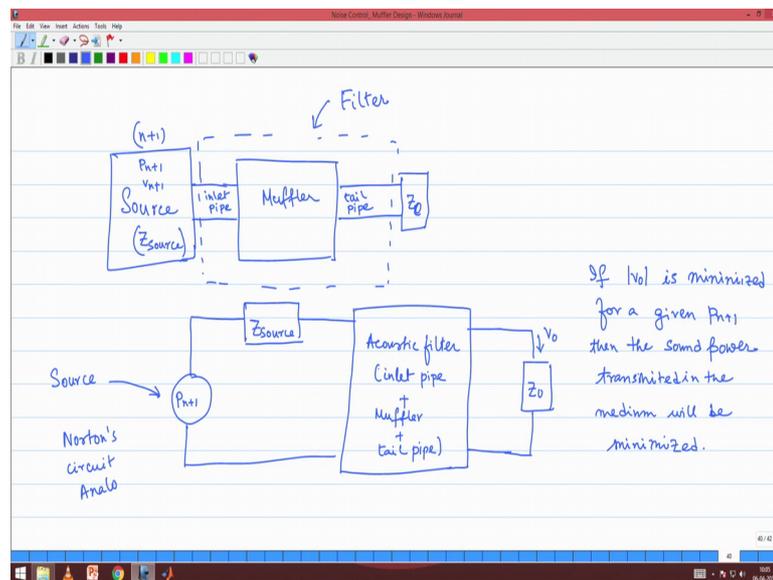


Acoustics & Noise Control
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Module – 27
Lecture – 32
Insertion Loss

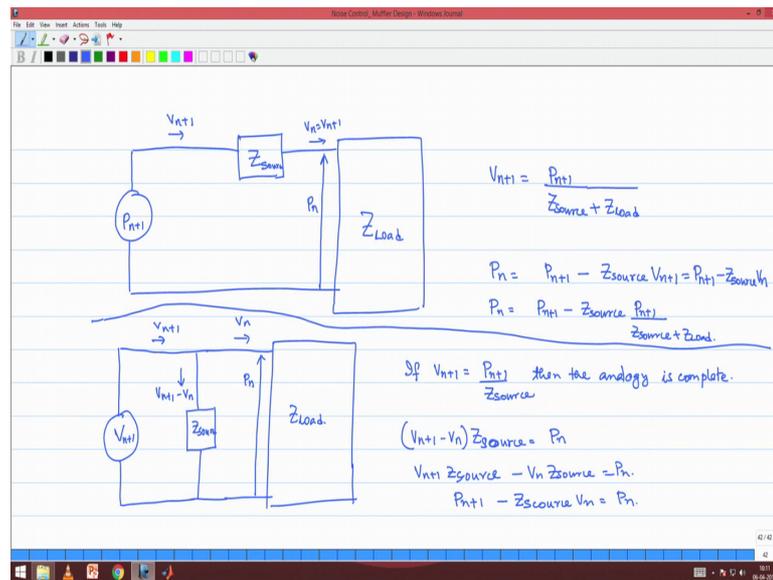
So, in the last class, we discussed the electro acoustic analogy and we also had some discussion about the source impedance. So, here was the (Refer Time: 00:28) and circuit that we had derived. So, here we had done the source modeling, the source was this P_{n1} plus 1.

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So, this is like saying that the source is a constant voltage source, but then that it goes through a source impedance and then reaches the load, right. So, this is what is called the Norton's circuit analogy. A completely equivalent picture is brought about by the Thevenin circuit analogy which is what will touch upon now.

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So, this is where in the source is model as a voltage source and then there is a source impedance Z_S or Z_{source} if you might call it and then there is a load. So, this load will include both the muffler as well as the radiation impedance. So, this part if you like is the load. So, I am just redrawing this picture by collating the muffler impedance together with the radiation impedance and calling it as the load impedance. So, in here what you will have is that though the source is supposedly working at a voltage of P_{n+1} , but what reaches the load is not P_{n+1} . This is P_n which is different from P_{n+1} because there is a source impedance which is connected to it.

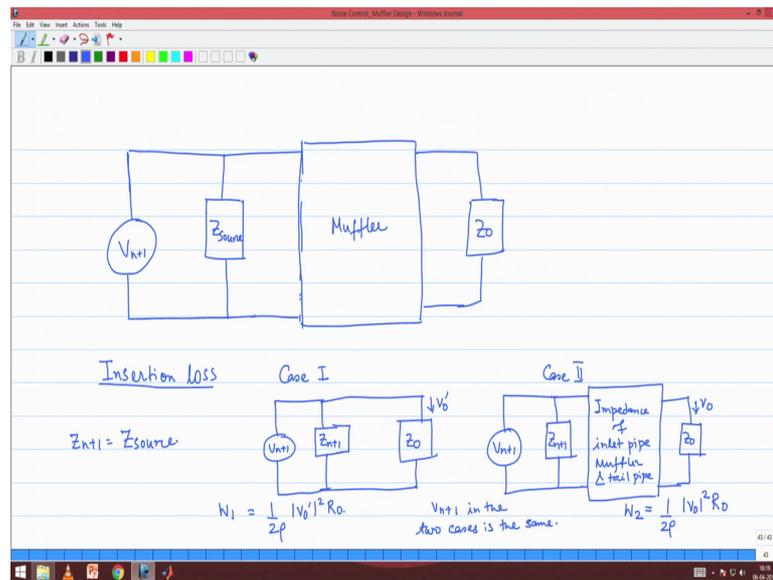
So, to calculate; what is the value of this P_n , we are doing this simple calculation. So, firstly, we understand that the current that is flowing through the circuit V_{n+1} is P_{n+1} divided by $Z_{source} + Z_{Load}$, right. So, therefore, P_n is $P_{n+1} - Z_{source} V_{n+1}$ which is also equals to V_n because between the input and the output terminals of the source impedance, there is no fall in current; that source impedance is in line, right, it is not in the shunt position. So, therefore, you could as well write this as P_n equals to $P_{n+1} - Z_{source} V_{n+1}$ divided by $Z_{source} + Z_{Load}$. Now we want to have another equivalent picture which will assume that we have a constant current source V_{n+1} to denote the current source. This is a constant current source and this current source is acting over the load, but this time since I am modeling the source as the current source rather than a voltage source, I have to give the source impedance in parallel.

So, let us see what this reveals and again we have the load impedance here, right. So, here the V_{n+1} will get split into 2 parts 1 is V_n . So, V_{n+1} and V_n will not be same in this situation because apart of the current will move in this shunt line which will be of the form $V_{n+1} - V_n$, right. So, if this is again P_n because we want this P_n to be the same we want to have an analogy between these 2 circuits and what we will show that if V_{n+1} is chosen as P_{n+1} divided by Z_{source} , then we will have then the analogy is complete. The analogy between the 2 source models; one in which we are modelling the source as a voltage source or in acoustic terms as a pressure source and the other in which we are modelling the source as a velocity source or in electrical terms as a current source.

So, let us see what our electrical modeling this time will give us, it will give us that $V_{n+1} - V_n$ into Z_{source} must be equals to P_n , right. So, in other words V_{n+1} into Z_{source} minus V_n into Z_{source} must be equals to P_n or the terminal voltage across these 2 ends. So, here what we have is $V_{n+1} Z_{source}$ will be gain P_{n+1} as per the definition. So, $P_{n+1} - Z_{source} V_n$ should be equals to P_n and here in this circuit, we had V_{n+1} to be the same as V_n itself. This is about; please understand the 2 circuits are different. So, in this circuit, we had V_{n+1} is equals to V_n . So, we could replace V_{n+1} here as V_n into Z_{source} into V_n . Now note that these 2 equations are exactly reading out to be the same with the condition that I have chosen my current source to be equal to the voltage source divided by the source impedance.

So, these 2 forms of source modeling are perfectly equivalent both in the electrical community as well as now that we are talking about the electro mechanical acoustic community, we will exploit this analogy of source modelling to give us better insights into how a muffler can be better designed. So, the point is this that if this condition V_{n+1} is equals to P_{n+1} divided by Z_{source} is invoked then we get to see that the current by applying the circuit loss in this second circuit, we are getting the same equation. These 2 equations are identical right which means that the 2 equations; the 2 circuits are completely equivalent.

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So, therefore, the velocity modelling of the source in our electro mechanical system would read in this following form, you will have a velocity source V_{n+1} and then you will have a source impedance which is in shunt position, then you will have the muffler which could have elements in line position which could have the elements in the shunt position and finally, you will have the radiation impedance Z_0 , right. So, this is the circuit when you do the modelling in terms of velocity source, whereas, what was shown earlier here is the circuit in terms of the source modeling taken as a pressure source, but both has its advantage. For example, if the source impedance is near 0, then what is happening is that the entire source voltage or the source pressure is getting communicated downstream where as if the source. So, in other words, if the source impedance is approaching 0, then this circuit is getting shorted, right.

So, if we wish to study the situation where source impedance is very low, then you can have this circuit to your help because that will be easier to interpret whereas, the interpretation for a very large source impedance can possibly be better with this modelling because when the source impedance is very large, this will be made an open circuit, see this will be a current source with an open circuit in the shunt position and that will have a very interesting special cases, but special cases apart; let us now try to define insertion loss which is a very important aspect in muffler design.

So, insertion loss; you consider 2 cases; case one when you have only the source and I am going to draw only the circuit, here you have only the source, you have the source impedance and you are having the radiation impedance for sure, Z_0 right and in case 2; I will call Z_{source} as Z_{n+1} because that will simplify my nomenclature Z_{n+1} is the source impedance. So, Z_{n+1} equals to Z_{source} and case 2 wherein, we will have a current source V_{n+1} a source impedance which is in the shunt position and then you have the impedance associated with the muffler and the inlet and exhaust pipes.

So, impedance of inlet pipe muffler and tail pipe; all of it goes in this box and finally, it comes out through the radiation impedance. In this 2 cases, we need to evaluate the 2 sound powers W_1 and W_2 sound power; in case 1 and sound power in case 2 has to be evaluated and then the difference between the 2 needs to be calculated for the insertion loss idea because this as I said is the real bench mark of how effective your muffler arrangement is. If you do not give any muffler or anything, it will have the value of the sound radiated to be W_1 , if you give a muffler, it has the value of the sound radiated to be sound power radiated to be W_2 , right the difference between the 2 is the insertion loss which is basically the benefit that you are getting by inserting the muffler together with its inlet and exhaust.

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Objective of muffler design would be to ensure minimal sound transmission into the atmosphere.

Sound power radiated into the atmosphere. Active intensity.

$$W = \frac{1}{2} \operatorname{Re} \left\{ p_0 v_0^* \right\} S$$

Recall $W = p S u$

$$= \frac{1}{2} \operatorname{Re} \left\{ p_0 \frac{v_0^*}{\rho s} \right\} S$$

$$= \frac{1}{2\rho} \operatorname{Re} \left\{ p_0 v_0^* \right\} = \frac{1}{2\rho} \operatorname{Re} \left\{ v_0 Z_0 v_0^* \right\}$$

$R_0 = \operatorname{Re} \{ Z_0 \}$ (resistance)

$$= \frac{1}{2\rho} |v_0|^2 R_0$$

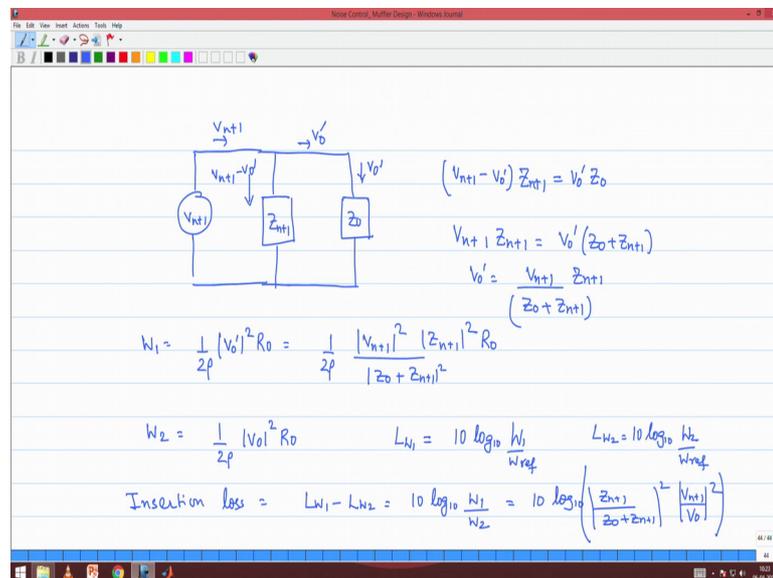
$\frac{p_0}{v_0} = Z_0$

So, that is the measure of the effectiveness of the exhaust system in terms of acoustic performance. So, here for W_2 ; we know that if V_0 is the current that was flowing

through the final load resistance we have already derived this expression that W_2 ; W will be $\frac{1}{2} \rho V_0^2 \text{ into } R_0$, right. So, this expression was already derived one by $\frac{1}{2} \rho \text{ mod of } V_0^2 \text{ square } R_0$, R_0 is the resistance associated with the radiation impedance value radiation impedance has reactants as well as resistance, we only care about the resistance, we do not care about the reactants while calculating the sound power because it is the time averaged power that is radiated out. Here also it is just the same, but the point is the V_0 , here even if the V_{n+1} , in both these cases is the same because the source is; obviously, the same even if the source is the same the amount of V_0 which goes through the impedance block Z_0 is going to be different. So, here it will be V_0 prime, right.

So, W_1 will be $\frac{1}{2} \rho V_0^{\prime 2} R_0$ please understand that V_{n+1} between the 2 system is same in the 2 cases is the same because it represents the same source Z_{n+1} is the same Z_0 is the same right between case 1 and case 2, but V_0 ; obviously, will be different because the amount of current which actually passes through the Z_0 impedance will be different depending upon what circuit you have in this plot.

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So, therefore, V_0 prime; we have to keep track of it. So, now, what we will do as a final reduction process is that we will try to relate V_0 prime in terms of V_{n+1} that should be trivial from the circuit analysis perspective. This is V_{n+1} , this is Z_{n+1} and this is Z_0 and we know that this is V_0 prime; this is V_{n+1} . So, what goes here is

also V_0' . So, what goes here is $V_{n+1} - V_0'$ right. So, $V_{n+1} - V_0'$ into Z_{n+1} is the potential drop across the first shunt across the impedance Z_{n+1} and that potential drop has to be messed with $V_0' Z_0$. So, this is the potential drop across this position right across the impedance Z_0 .

So, therefore, what you have is $V_{n+1} Z_{n+1} = V_0' Z_0 + Z_{n+1}$ and finally, V_0' is equals to V_{n+1} divided by $Z_0 + Z_{n+1}$ into Z_{n+1} right. So, therefore, W_1 which was the sound power which is radiated in the situation where there is no muffler no inlet pipe no tail pipe nothing of that sort. So, W_1 was given as $\frac{1}{2} \rho V_0'^2 R_0$ and V_0' ; we wish to now write it as $\frac{1}{2} \rho V_{n+1}^2$ divided by $\text{mod of } Z_0 + Z_{n+1}$ whole square divided by $\text{mod of } Z_{n+1}$ whole square. So, this is the expression for the sound power that is radiated this is the sound power radiated in the linear scale will come to the log scale soon enough, but this is just the sound power that is radiated or power that is lost in the circuit.

Student: Into R.

Into R_0 yeah into R_0 ; I have missed out. So, this is the sound power that is radiated in the case where there is no muffler and W_2 is going to be any way $\frac{1}{2} \rho \text{mod } V_0^2 R_0$ then what we have to do is we have to find the sound power in the logarithmic scale. So, that should be easy sound power in the logarithmic scale as we know is $10 \log_{10} W_1$ by $W_{\text{reference}}$ value and similarly $L W_2$ is going to be $10 \log_{10} W_2$ by W_{ref} .

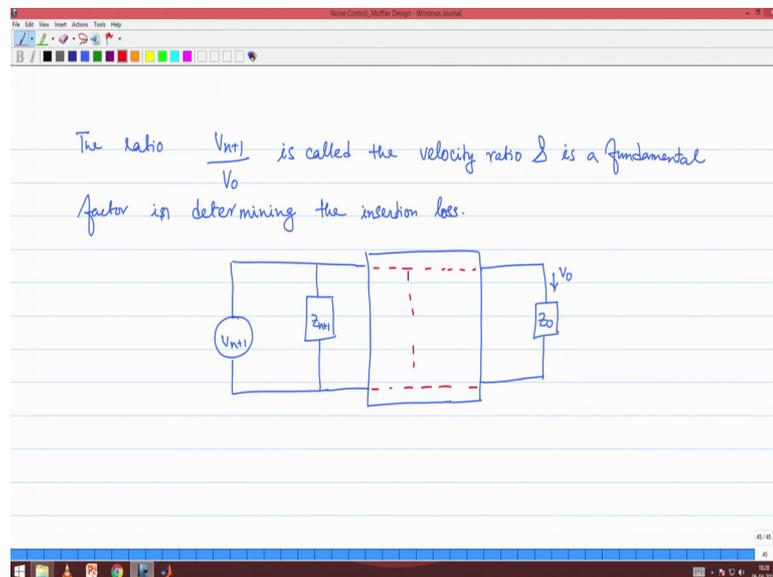
So, therefore, the insertion loss is going to be $L W_1 - L W_2$ which is $10 \log_{10} W_1$ by W_2 and then if we put the appropriate values now which we have calculated for W_1 and W_2 , it reads in the following fashion the R_0 term. Now you see it actually does not affect the insertion loss the R_0 term you see is going to get cancel from both the numerator and the denominator in that ratio; what we have will be left with is Z_{n+1} divided by $Z_0 + Z_{n+1}$ magnitude square into V_{n+1} divided by V_0 magnitude square this is the insertion loss formula, right.

Now; obviously, Z_{n+1} and Z_0 are sort of something for which you do not have any control Z_0 is the radiation impedance it is god given, we have to just accept it, similarly source impedance depends upon the mechanical or the fluid dynamic properties of the

source you cannot change it. So, the only way in which you can change the insertion loss to your benefit is to look at this ratio of V_{n+1} plus $1/V_0$.

Let us understand; what is this ratio V_{n+1} by V_0 . So, if you look at this circuit diagram, it is very clear that from the current at the source which is V_{n+1} only a part of it will be able to reach the load which is of the value V_0 . So, if you have some lines which are in the parallel position and divert the current not towards V_0 , but towards the other parallel bypass lines then you will achieve a very large reduction of V_0 with respect to V_{n+1} .

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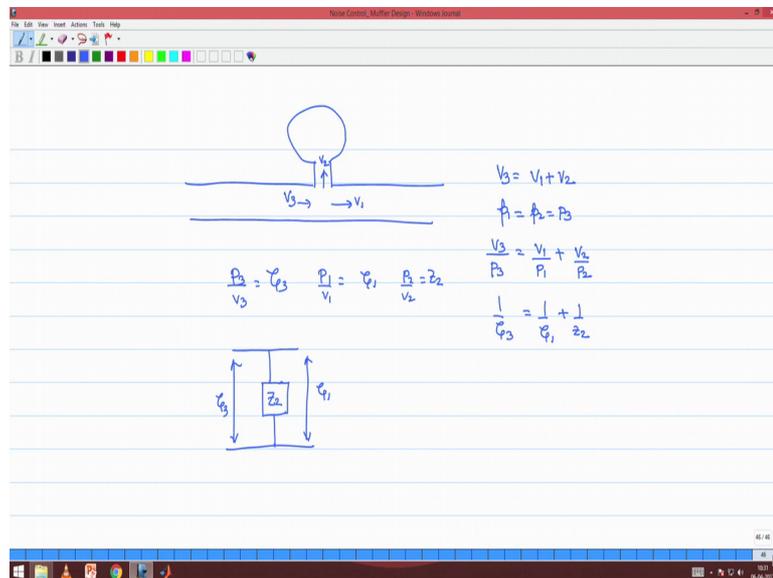
So, this ratio is of critical importance the ratio V_{n+1} divided by V_0 is called the velocity ratio and is a fundamental factor in determining the insertion loss and insertion loss as I said it is just the benefit that you get by addition of your exhaust system, if you do not have an exhaust system and you let the source, directly radiate out into the atmosphere then you will get a sound power radiated as W_1 and on the other side it is W_2 , if you have an exhaust system, if you just take a difference between these 2 sound powers in the logarithmic scale this is what is insertion loss. So, insertion loss is the true benchmark for the performance of the acoustic performance of your exhaust system.

So, what it turns out here; from here what it turns out is the following you have to have some bypass paths in this block such that most of the current does not flow through this path, right, you have to reduce V_0 for a certain value of V_{n+1} , how can you do

that? If you have a certain within this block; if you have a short circuit path; let us say that will be ideal if you have a short circuit path of this sort; what will happen, then the entire current will go; throw this short circuit path and minimal current will achieve, go through the radiation impedance Z_0 and therefore, V_0 will be drastically less and V_0 plus 1 and V_0 that ratio will be very high insertion loss will very high.

So, that is how we should go about trying to design for good insertion loss for a muffler, right, sometimes the problem is you will not have to; in your table the knowledge of the source impedance because let us say you are a vendor for just the muffler component and there is no way that you know the source impedance in which case transmission loss may be sort of the best approximation that you can do to bench mark your mufflers. But once you know that this muffler has to be fitted with a particular source, then you must always calculate the insertion loss and to choose the best muffler for a given source you must do the insertion loss calculation unless you do the insertion loss calculation based on transmission loss the exercise could go completely futile.

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So, let us quickly illustrate how you can create a bypass path. So, suppose now we have an exhaust system which is this form. So, what I have done is I have put a Helmholtz resonator on my exhaust pipe. So, what is going to happen here? So, here I have an incoming velocity, I mean acoustic velocity V_3 and there here there is an outgoing acoustic velocity V_1 and here there is also an outgoing acoustic velocity V_2 , right. So,

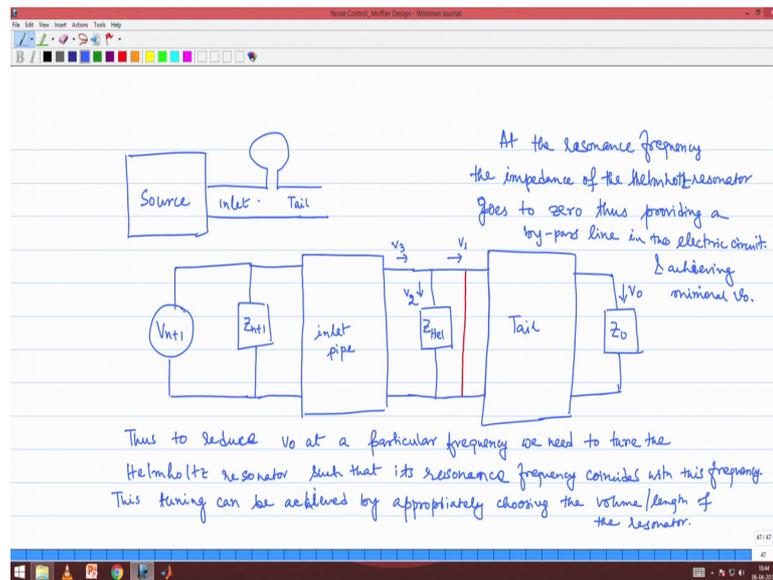
applying conservation of mass flow at these junction point we get V_3 is equals to V_1 plus V_2 , right, what about the pressures because all these points are continuous all the points will have the same pressures P_1 equals to P_2 is equals to P_3 , right. So, now, if this is so then what we can write is V_3 by P_3 has to be equals to V_1 by P_1 plus V_2 by P_2 if we call the impedance P_3 by V_3 as ζ_3 then this can be written as 1 by ζ_3 ; similarly if we call P_1 by V_1 impedance as ζ_1 then we could write this as ζ_1 and finally, if we call P_2 by V_2 as the impedance Z_2 then we could write this as Z_2 by Z_2 .

So, in this situation we are going to have if you just look at what happens upstream point of the Helmholtz resonator and what happens at the downstream point of the Helmholtz resonator you are going to find that V_3 is going to get bifurcated or sub divided into 2 parts right and V_3 stands for the current the velocity stands for the current in our analogy. So, the current is going to get divided the voltages are going to remain same which is the pressures and the impedance that you are now seeing is effectively that of the equivalent circuit of in which you have parallel impedances in parallel, right.

So, the equivalence circuit for this case would be as simple as this, you have 1 impedance Z_2 which is in parallel and this impedance is ζ_3 and this impedance is ζ_1 . So, ζ_3 impedance can be recovered from ζ_1 impedance and from Z_2 by simply appealing to the impedance of 2 parallel circuits, right and with this again what you can find out is that if the Z_2 impedance and what is the Z_2 impedance Z_2 impedance is exactly the impedance of the Helmholtz resonator because V_2 is exactly the impedance at the inlet of the Helmholtz resonator, right.

So, if the impedance of the Helmholtz resonator goes to 0, what will happen this path will be short circuited, right? So, this is the circuit of only this portion, now if I add a source. So, in the next picture I will add a source.

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And I will add a Helmholtz resonator and a tail pipe which opens into the open atmosphere right. So, this is the source and this is the Helmholtz resonator. So, what will happen here the equivalent electrical circuit will read in the following fashion you have a V_{n+1} associated with the source you have a Z_{n+1} associated with the source impedance, you have a certain tail pipe sorry, an inlet pipe. So, this is the impedance associated with the inlet pipe and remember the pipe impedance we have discussed in the form of sorry the pipe modelling we have done it in the form of transfer matrix in general it is neither in shunt position nor in inline position it is a distributed impedance.

So, we should not put the impedance associated with the inlet pipe in either inline position or shunt position, but we may put it in this form to denote that it is a distributed impedance something which cannot be put in the 2 extreme cases, then we should have a Helmholtz resonator and the Helmholtz resonator will be in parallel as we have discussed. So, this I will call as Z_{HCL} associated with $Z_{Helmholtz}$ and then you will have the tail pipe, this is the tail pipe, this is the tail pipe, this is the inlet pipe and that is the Helmholtz resonator and finally, it has to open to a radiation impedance set 0, right.

Now, the point is this that out of all these, if you know that you are hearing a large sound at particular frequencies, then what should you do? You should tune the Helmholtz resonator; remember the Helmholtz resonator has resonance frequency, we have described that the mass effect is there, the spring effect is there, one of it is V by c square

the other of it is L by S you can tune the frequencies associated with this Helmholtz resonators such that the problematic frequencies are exactly the resonance frequencies of your Helmholtz resonator. In that case, what happens if the frequency happens to be the resonance frequency of the Helmholtz resonator? What will happen this part of the circuit will show 0 impedance, right, 0 impedance is the situation where you have resonance frequency, once you have resonance frequency, this impedance will turn 0 and then it will be replaced by a short circuit line and then in the electrical terminology.

Now it is very easy to understand once this line is shorted, most of the current will flow through this line it will not prefer to go anywhere else right and as a result the V_0 which passes through remember your ultimate objective is to minimize V_0 that V_0 will get revised.

Student: Here we are during the Helmholtz resonator for only one frequency.

Yes.

Student: So, the rest of the frequency will still pass.

Yes, this is I have now killed one the most problematic frequency in my frequency spectrum. I would see that there are lots of frequencies which are contributing, right, but I can always find out which is the most dominant frequency contributor, right, I will kill 1 with 1 Helmholtz resonator. Now if I have multiple frequencies, I will use multiple Helmholtz resonators, I can give as many Helmholtz resonators as I want in this pipe and start knocking of each of this frequencies, it is a very scientific method that I at least you understand that you understand, how you can knock off one, right, if you know how you can knock of one you can knock off the others also by using a similar strategy for all the other frequencies, but the point I wish to drive home with this electro mechanical analogy which is very difficult to do by staying in mechanical systems or even in acoustic systems is that this idea of shorting the line the Helmholtz resonator is actually providing a short circuit to the equivalent electrical system and thereby saving the current to go into the radiation impedance.

Student: Sir, more enters the Helmholtz resonator is it coming back in to the circuit.

No, all that is taken care of see; what we are saying is that from here on what goes ahead is in that terminology we had said V_1 and what goes here is V_3 and what goes here is sorry this is V_2 and this is V_3 , right. So, from here on what goes from downstream of the Helmholtz resonator is only V_1 , right, downstream of the Helmholtz resonator this in the previous diagram, if you see the downstream of the Helmholtz resonator is V_1 . See here what is happening is a standing wave is getting generated it is not that it is only a travelling wave and you get reflected a standing wave situation has already create that is why it is a resonator, right otherwise you cannot have a resonator a standing wave situation has created its only that now we are appealing to the impedance of the Helmholtz resonator and we are tuning the resonance frequencies such that you have a 0 impedance in the bypass line.

Thus to reduce V_0 at a particular frequency we need to tune the Helmholtz resonator such that its resonance frequency coincides with this frequency this tuning can be achieved by appropriately choosing the volume or the length of the resonator.

So, I think just one more; note I will do before I go today that is at the resonance frequency the impedance of the Helmholtz resonator goes to 0 thus providing a bypass line in the electric circuit and achieving minimal V_0 . So, if your V_0 is minimized velocity ratio will be increased and your insertion loss will increase. So, that is the demonstration of the insertion loss of again a pretty much academic muffler I would say, but quickly in the next class, we will show some practical construction where this idea of a Helmholtz resonator will get nicely approximated.

So, till then good bye.