

Course on Principles of Communication Systems-Part 1

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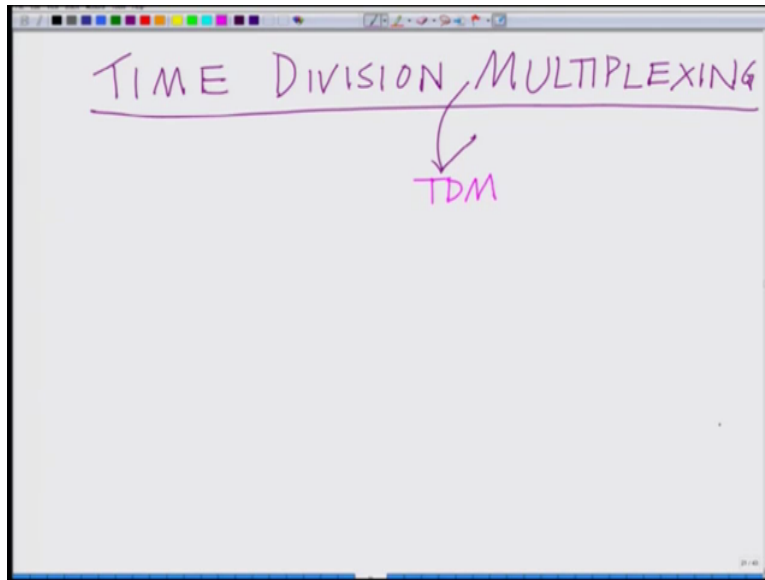
Lecture 52

Module 8

Bandwidth requirements of Time Division Multiplexing (TDM), The T1 TDM System: A Case Study

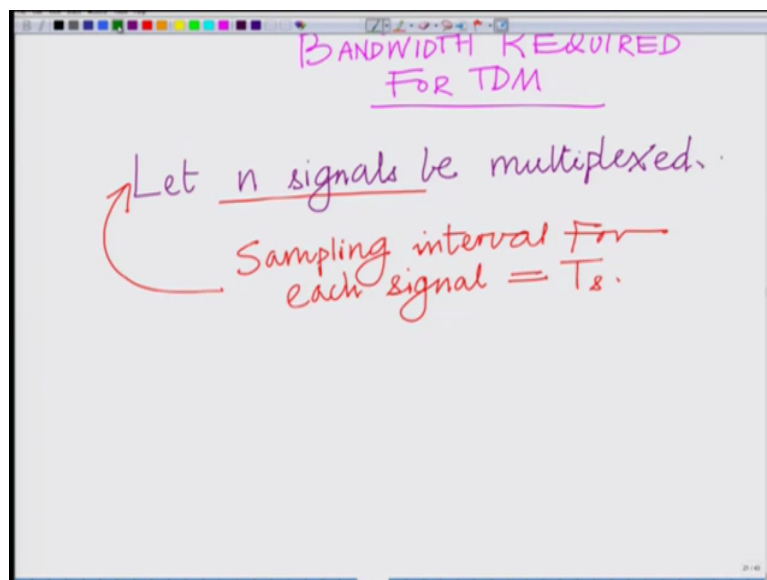
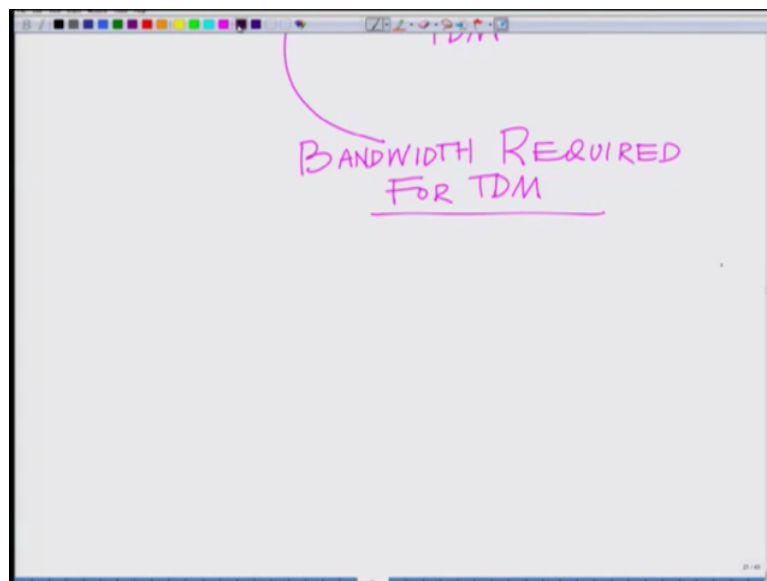
Hello, welcome to another module in this massive open online course alright. so we are looking at Time Division Multiplexing alright, so let us look at other aspects of Time Division Multiplexing in this module, ok.

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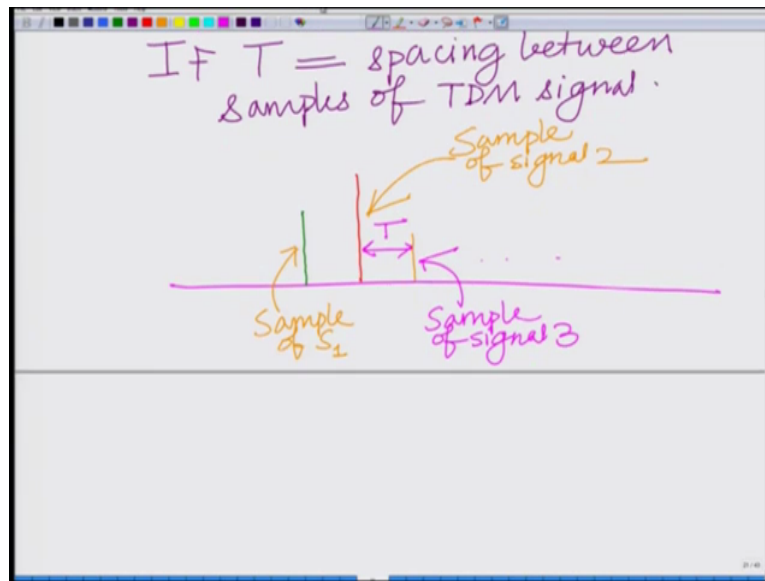
So we are looking at Time Division Multiplexing that is TDM technology to multiplex several signals over a single channel for transmission ok. So this is your Time Division Multiplexing technology.

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Now let us start by looking the bandwidth required for TDM, so what we want to look at is first we want to answer the question what is the bandwidth what is the bandwidth required for TDM. Now if you look at this for instance let us say let there be n signals so let us say we are multiplexing n signals using Time Division Multiplexing, so let n signals be multiplexed and they are sampled with all are sampled with at equal intervals T_s so sampling interval for each signal equals T_s , ok. So there are n signals and the sampling interval for each signal is T_s .

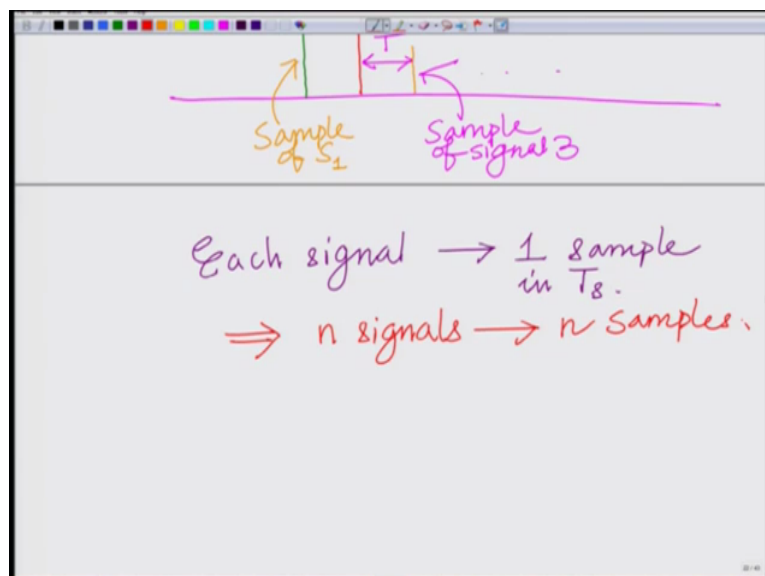
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Now if T is the spacing between the samples if T equal to the denotes the spacing between samples of your TDM signal that is if you look at your TDM signal correct and let say we look at the samples of the multiplex signals. So this is sample of signal 1, sample of signal 2, sample of so this is sample of signal this is samples of the multiplex signal. So sample of signal 1, sample of signal 2, this is sample of signal 3, sample of signal 3, so on and so forth and if this duration the duration between samples is T .

Now we have to note that the sampling duration of each signal is uniformed that is T_s , which means every signal has a single sample in T_s . Therefore n signals will have n samples in the interval T_s , right.

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Each signal $\rightarrow \frac{1}{n}$ sample in T_s .

$\Rightarrow n$ signals $\rightarrow n$ samples.

Spacing between samples of TDM signal $= T$

$$= \frac{T_s}{n} = T$$

So each signal contributes 1 sample in T_s , implies when you look at n signals. So naturally they are going to give n samples, therefore distance between therefore the spacing between samples spacing between samples, since in T_s there are n samples spacing between samples of your TDM signal equals T_s divided by n that is equal to T . So sample spacing equals spacing between samples equals of TDM signal equals T and we have T equal to T_s divided by n .

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BW required for Transmission of TDM signal.

$$= \frac{1}{2T} = \frac{1}{2T_s/n}$$

$$= \frac{n}{T_s}$$

$$\begin{aligned}
 &= \frac{1}{2T} = \frac{1}{2T_s/n} \\
 &= \frac{n}{2T_s} \\
 &= \frac{1}{2} n F_s \\
 &F_s = \frac{1}{T_s} \\
 &= \text{Sampling Freq.}
 \end{aligned}$$

Now bandwidth that is required for transmission of TDM signal bandwidth required for transmission of TDM signal this is equal to well the bandwidth is 1 over 2T which is equal to well we have seen T equals Ts divided by n, so this is 1 over 2Ts divided by n, which is n over Ts, now n sorry this is n over twice Ts which is half nFs where Fs equals the sampling frequency Fs equals 1 over Ts equals the sampling frequency, ok so Fs equals 1 over Ts which is equal to the sampling frequency.

Now if we sample at the Nyquist rate then if we sample the signals at the Nyquist rate, right by the Nyquist Criterion the sampling frequency Fs has to be greater than atleast 2Fm where Fm is the maximum frequency of each signal, ok.

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$$\begin{aligned}
 &F_s = \frac{1}{T_s} \\
 &= \text{Sampling Freq.}
 \end{aligned}$$

From Nyquist Criterion,

$$F_s \geq 2F_m$$

BW required for Transmission of TDM signal.

$$f_{TDM} = \frac{1}{2T} = \frac{1}{2T_s/n}$$

$$= \frac{n}{2T_s}$$

$$= \frac{1}{2} n f_s$$

$f_s = \frac{1}{T_s}$
= Sampling freq.

$$f_s \geq 2f_m$$

$$f_{TDM} = \frac{1}{2} \cdot n \cdot f_s$$

$$\geq \frac{1}{2} \cdot n \cdot 2f_m$$

$$= n f_m$$

So by Nyquist Criterion or from Nyquist Criterion from the Nyquist Criterion we need f_s greater or equal to $2f_m$ which means bandwidth, therefore if we call this bandwidth required for transmission as f of TDM we need f of TDM f of TDM equals well half n into f_s but f_s greater than equal to $2f_m$, therefore this is greater than the half n into twice f_m which is equal to n times f_m .

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$$F_{TDM} \geq n F_m$$

= # signals multiplexed,

Maximum Frequency of each signal

Therefore we need the bandwidth of the Time Division Multiplexed signal to be greater than equal to n times F_m , where F_m is the well so F_m so this is n , this is equal to number of signals multiplexed and F_m is the maximum frequency of the multiplex signals, this is equal to maximum frequency component of each signal, ok maximum frequency component of each signal, ok.

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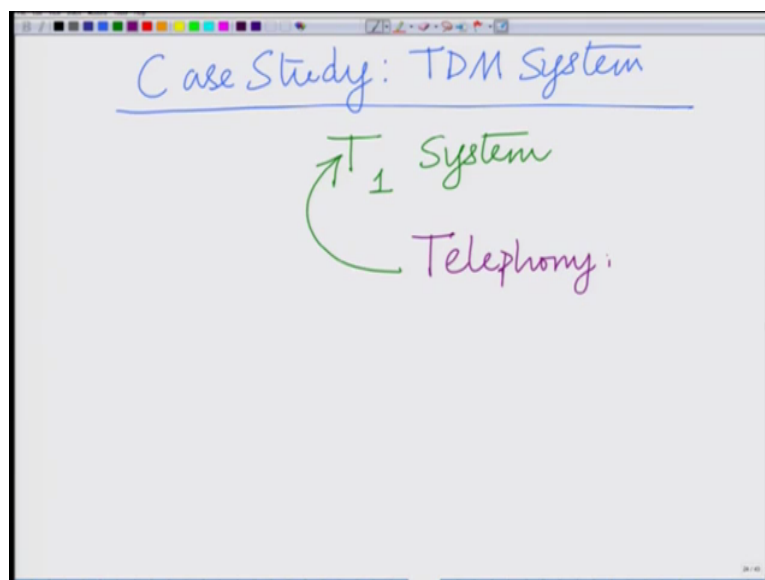
$$\text{BW of TDM Signal} \geq \# \text{ signals multiplexed} \times \text{BW of each signal.}$$

So basically the bandwidth of you can say bandwidth of TDM signal is greater than or equal to number of signals multiplexed the product of number of signals multiplexed into maximum frequency, that is or you can say bandwidth of each signal times times the bandwidth of each signal. So that is the relation that we have for the bandwidth of TDM

signals. So the TDM signal ofcourse naturally since it is multiplexing a large number of signals we required a large bandwidth, right. This bandwidth is related to the sampling frequency which is in term related to the bandwidth from the Nyquist Criterion related to the bandwidth of each signal.

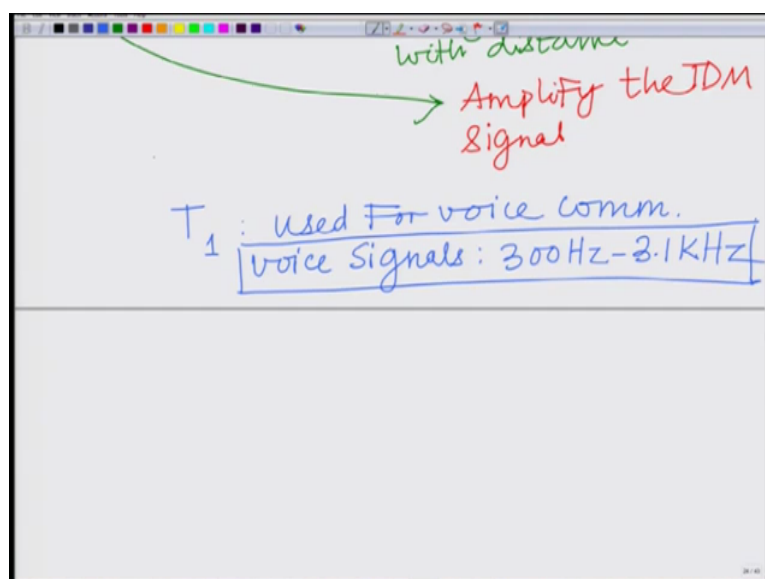
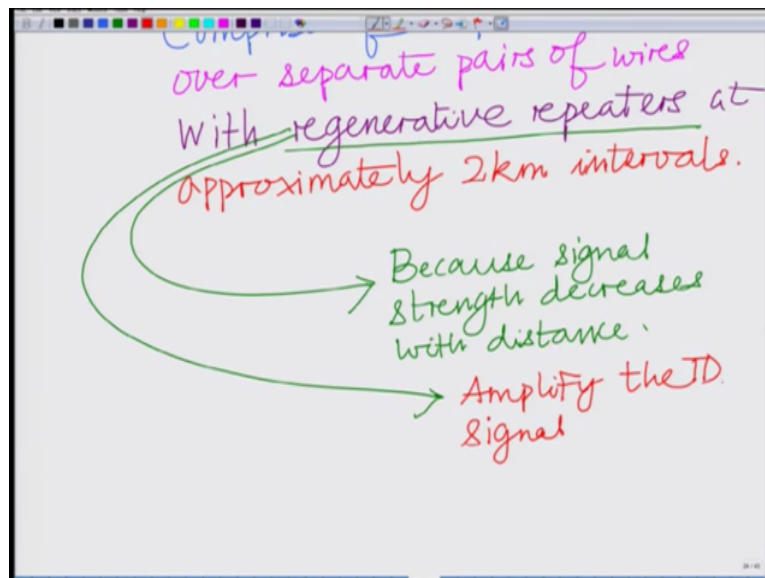
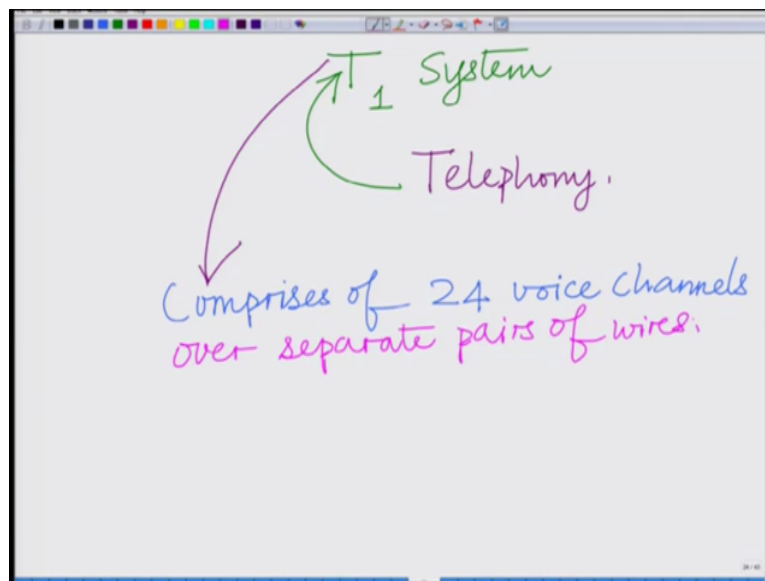
So what we have shown is the bandwidth FTDM of this Time Division Multiplex signal has to be atleast n times the bandwidth of the signals being multiplexed, ok. So it requires a large bandwidth for transmission, ok. Now let us do a simple look at a simple case study to understand that construction and working of a typical TDM signal that is a typical TDM system that is a Time Division Multiplex system also known as a T1 system, ok.

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So now what we want to do is we want to do a case study for TDM system in particular we want to consider one particular TDM system that is the that is the T1 system that is used to a telephone. This is used to a voice communication voice telephone, ok so this is used for telephone, ok.

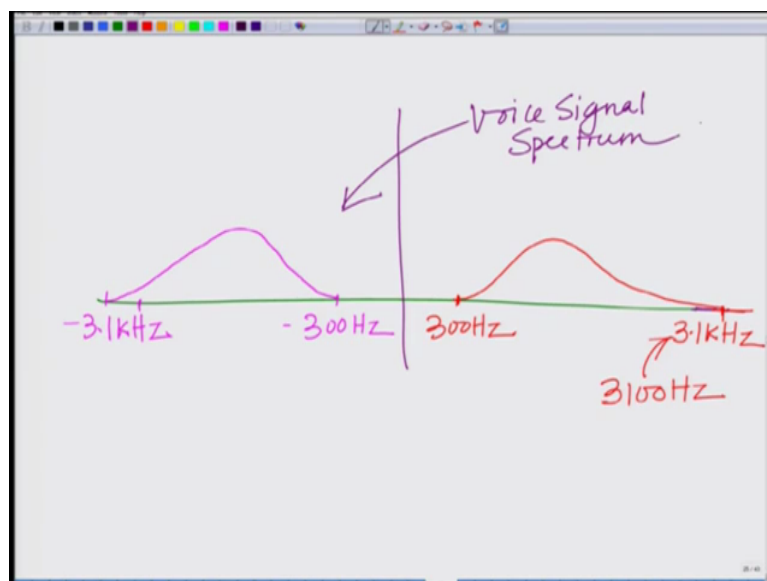
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The TDM system this comprises of 24, now the T1 system is comprises of 24 is comprises of 24 comprises of 24 voice channels over separate pairs of wires over over separate pairs of wires with regenerated repeaters because the signal dies down with distance separate pairs of wires which basically amplify with regenerative repeaters at approximately 2 kilometer intervals, ok and why do we need this regenerative repeaters we need the regenerative repeaters to repeat the signal because signal strength decreases with distance because the signal strength decreases with distance, ok.

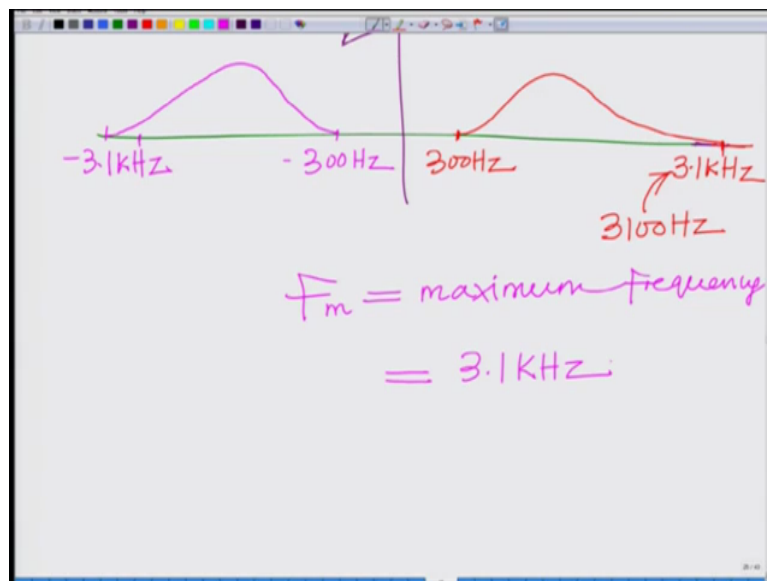
So these regenerative repeaters what they do is they amplify the signal regenerate that is prevent the signal from dying down see they amplify the the signal amplify the TDM signal, ok. So this regenerative repeaters amplify the TDM signal and this is used for voice T1 used for voice communication and voice signals have a frequency range that is 300 hertz to so the voice signal frequency band is 300 hertz to 3.1 kilohertz, ok.

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So what you have is basically you have a voice spectrum that looks something like you have a voice spectrum that looks something like this so it is the start set your 300 hertz ended something like start set 300 hertz and goes upto 3.1 kilohertz that is 3100 hertz. Similarly if you can look at this it starts with this and goes upto well right. So this is your voice spectrum is typically starts at your minus ok. So this is your voice spectrum so it start minus (300) minus 3.1 kilohertz to minus 300 hertz, this is your how your typical voice spectrum looks like or voice signal this is the spectrum of the voice signal, ok.

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Now therefore you can see the maximum frequency F_m that is your maximum frequency in this case study of the voice signal we are consider as the multiplexing of 24 voice signal, maximum frequency equals well that is 3.1 kilohertz that is the human frequency maximum frequency of human voice signal is 3.1 kilohertz, ok.

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$F_m = \text{maximum frequency}$
 $= 3.1 \text{ kHz}$

From Nyquist Criterion
To avoid Distortion

$F_s \geq 2F_m = 2 \times 3.1 \text{ kHz}$
 $= 6.2 \text{ kHz}$

Handwritten whiteboard content:

$$F_s \geq 2F_m = 2 \times 3.1 \text{ KHz} = 6.2 \text{ KHz}$$

$$F_s \geq 6.2 \text{ KHz}$$

Nyquist Sampling rate
To avoid aliasing

Now therefore the sampling frequency F_s from Nyquist Criterion so from Nyquist Criterion to avoid distortion to avoid distortion we need F_s greater than or equal to twice F_m which is twice into 3.1 kilohertz that is equal to 6.2 kilohertz, ok. So we need 6.2 kilohertz. However F_s greater than or equal to so we need minimum F_s is basically 6.2 kilohertz that is to avoid aliasing we know this right, this is basically the Nyquist Nyquist sampling rate minimum required sampling rate this is Nyquist sampling rate to avoid to avoid aliasing, ok.

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Nyquist Sampling rate
To avoid aliasing

Typically sampled at

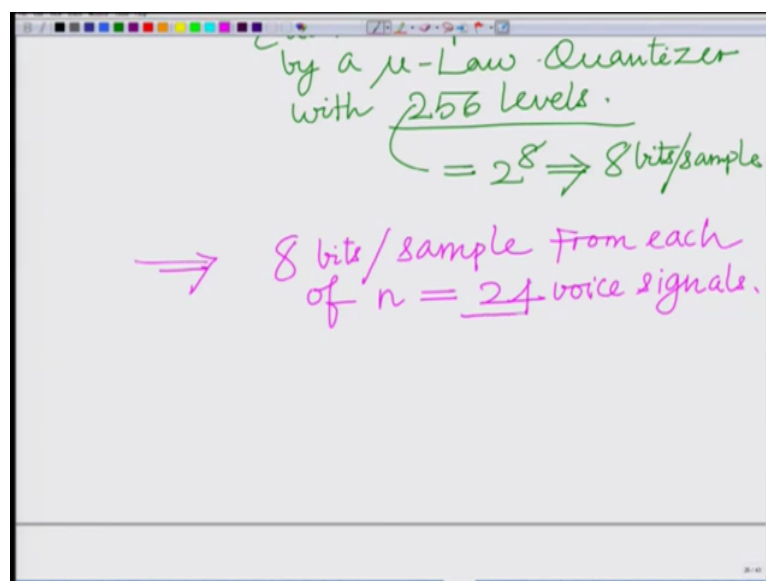
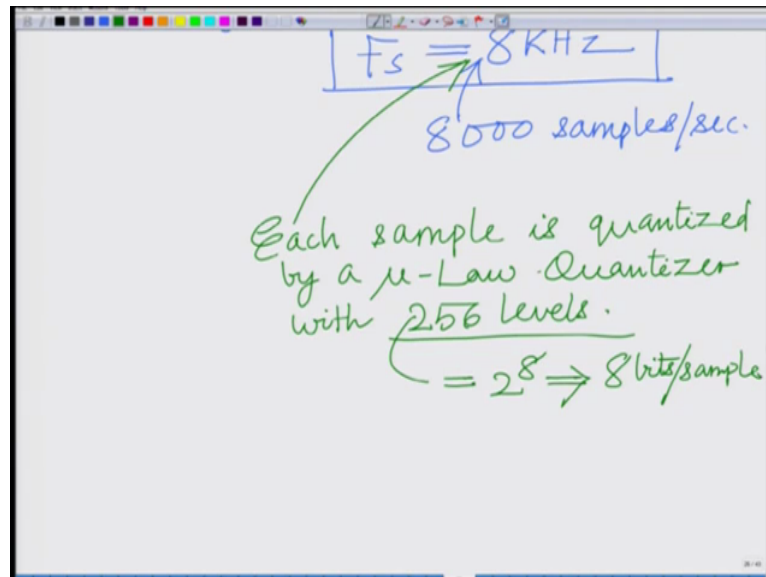
$$F_s = 8 \text{ KHz}$$

8000 samples/sec.

Now this is quantized with a mu law quantizer, however it is typical. So even though 6.2 kilohertz is sufficient it is typically sampled at 8 kilohertz at F_s equal to 8 kilohertz that is 8 kilo samples per second ok 8 kilo sampled at 8 kilohertz means 8 or 8000, 8 kilo means 8000 samples per second. So even though the minimum required sampling rate is 6.2 kilohertz, it is

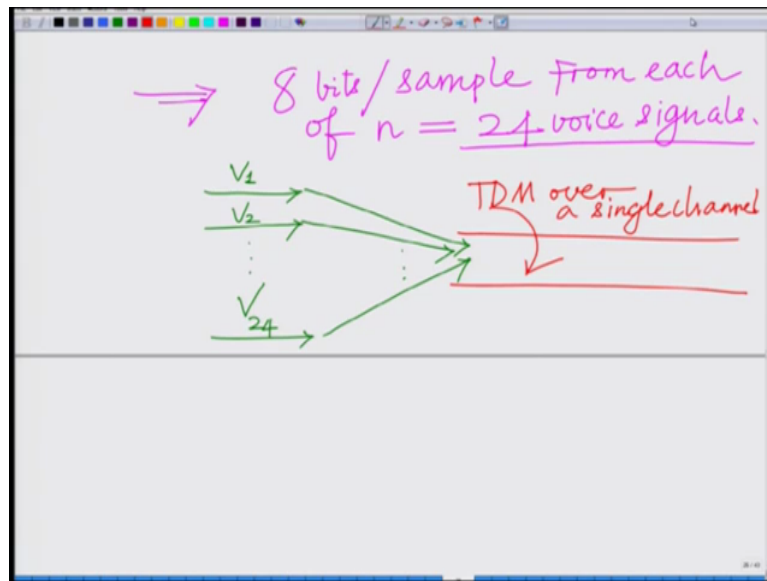
over sampled right it is sampled at a slightly higher rate it is sampled at 8 kilohertz that is 8 kilo samples or 8000 samples per second.

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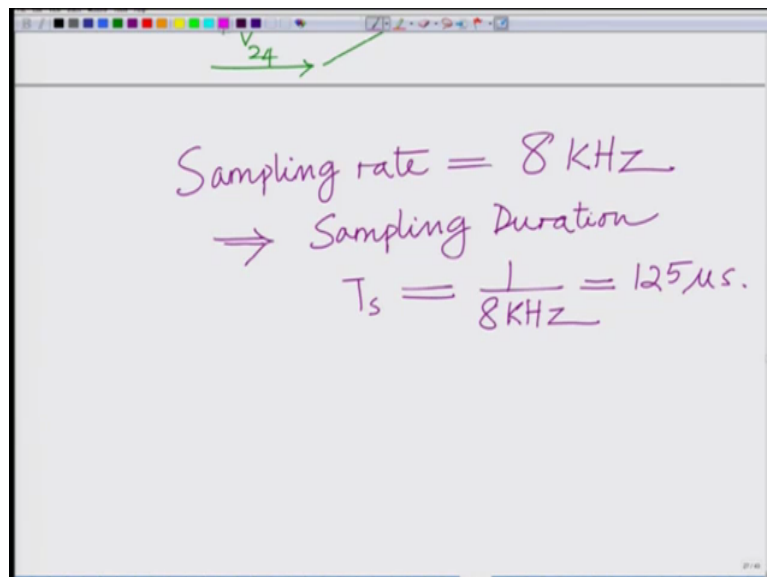
Now each sample is quantized using 8 bits alright so we have well what do we have we have each sample is quantized by a mu law quantizer using 8 bits, ok with 8 bits mu law quantizer with 256 levels where 256 equals 2 to the power of 8 implies 8 bits per sample this implies 8 bits per sample therefore we have 24 voice signals right 8 kilohertz sampling rate so net system we have is 8 bits per sample from each of n equal to 24 n equal to 24 voice signals that is your T1 system.

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So what you have in your T1 system is basically if you look at this you have well you have signal 1, signal 2 24 so you have V_1, V_2 , so on upto V_{24} all these are multiplexed over a single channel. So all these are TDM Time Division Multiplex over a over a single channel, ok.

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⇒ Sampling Duration

$$T_s = \frac{1}{8 \text{ KHz}} = 125 \mu\text{s.}$$

Frame Duration

Each of 24 signals generates 1 samp

So naturally what is the well sampling rate equal to 8 kilohertz 8 kilohertz implies sampling duration equals well 1 over 8 kilohertz T_s equal to 1 over 8 kilohertz that is equal to 125 micro seconds, ok. Now each now this we can also call this as frame duration we can also call this as frame duration, ok. In this basically each signal generates so in this each signal generates 1 sample, each of the 24 signals generates 1 sample in a frame duration each of 24 signals generates 1 sample 1 sample each sample is quantized using 24 bits.

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Frame Duration

Each of 24 signals generates 1 sample.

$$\begin{aligned} \# \text{ bits in frame} &= 24 \times 8 \text{ bits} \\ &\quad + 1 \text{ sync pulse} \\ &= 192 + 1 \\ &= 193 \text{ bits.} \end{aligned}$$

So in a frame duration you have that is number of samples in frame or you can say number of bits in frame equals well 24 number of signals into well 1 sample per signal into 8 bits per sample. So 8 bits from the (24) so 24 signals each generates a sample with 8 bits. So 24 into 8

plus 1 synchronization pulse or synchronization bit. So that is equal to 24 into 8 so 192 plus 1 synchronization 1 synch bit that is equal to 193 bits, total bits per frame equals 193.

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Handwritten notes on a digital whiteboard. At the top right, there is a small diagram of a pulse labeled "T1 pulse". Below it, the calculations are as follows:

$$\begin{aligned} &= 192 + 1 \\ &= 193 \text{ bits.} \end{aligned}$$

In T₁ system, # bits/Frame = 193

⇒ Time duration between bits = 125 μs.

Handwritten notes on a digital whiteboard. The calculations are as follows:

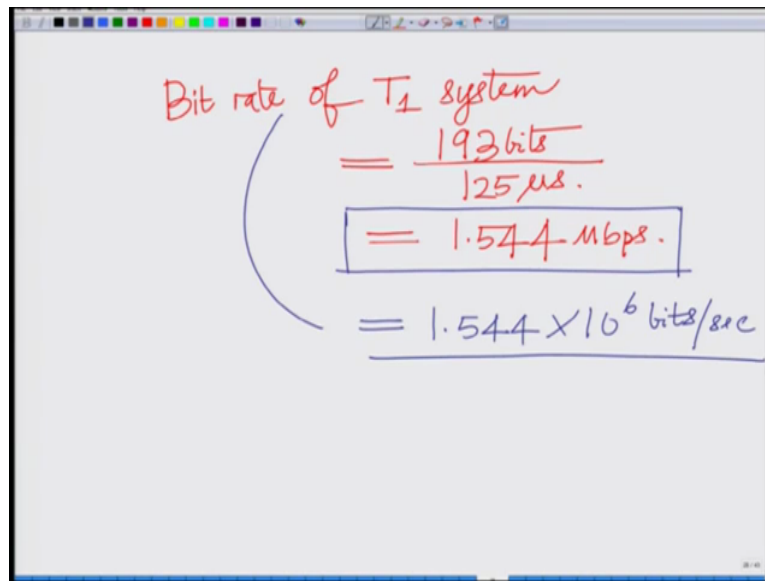
In T₁ system, # bits/Frame = 193

⇒ Time duration between bits = $\frac{125 \mu s.}{193 \text{ bits}}$

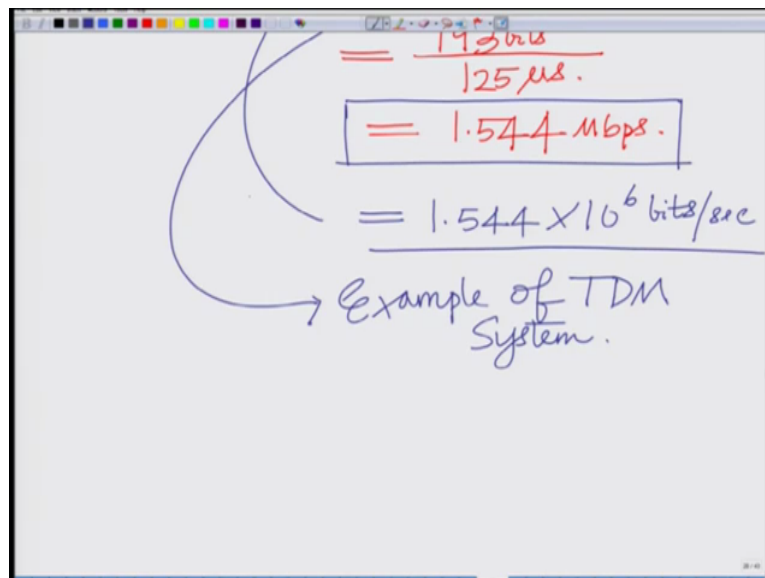
$= 0.647 \mu s.$

So in this TDM T1 system you can see in the TDM system in the case study that we are doing number of bits per frame equal to 193, which implies well duration of each bit or time interval between bits frame duration is 125 micro seconds, time duration between bits equals 125 micro seconds, ok. The time duration between bits is 125 micro seconds divided by well 193 bits 193 bits that is equal to well 0.647 micro seconds.

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Handwritten calculation of the bit rate of a T1 system. The text 'Bit rate of T1 system' is written in red. Below it, the calculation is shown in red ink:
$$= \frac{193 \text{ bits}}{125 \mu\text{s.}}$$
 This result is boxed in blue. Below the box, the calculation continues in blue ink:
$$= 1.544 \text{ Mbps.}$$
 Finally, it is written in blue ink:
$$= 1.544 \times 10^6 \text{ bits/sec}$$
 A blue curved line connects the boxed result to the final expression.



Handwritten calculation of the bit rate of a T1 system, identical to the one above. The text 'Bit rate of T1 system' is written in red. Below it, the calculation is shown in red ink:
$$= \frac{193 \text{ bits}}{125 \mu\text{s.}}$$
 This result is boxed in blue. Below the box, the calculation continues in blue ink:
$$= 1.544 \text{ Mbps.}$$
 Finally, it is written in blue ink:
$$= 1.544 \times 10^6 \text{ bits/sec}$$
 A blue curved line connects the boxed result to the final expression. Below the final expression, the text 'Example of TDM System.' is written in blue ink, with an arrow pointing from the boxed result to it.

And bit rate of the T1 system bit rate of the T1 system is basically 193 bits divided by 125 micro seconds that is equal to well 1.544 megabits per second. So this is basically the bit rate of the T1 system if you calculate that that is a huge bit rate that is equal to 1.544 megabits that is into 10 to the power of 6 bits per second that is your bit rate of the T1 system, ok. And the T1 system is a classic example of a Time Division Multiplexing, this is a classic example of a TDM system therefore what we have is what we have seen in this module is you have seen some important aspects of the TDM system that is what are the bandwidth requirements of the TDM system also we have done a case study of a classic example of a TDM system that is the T1 system which combines 24 digital voice channels into a single system right and we have seen that this has a huge bit rate of 1.544 approximate 1.544 megabits per second, alright so we will stop this module here thank you.