

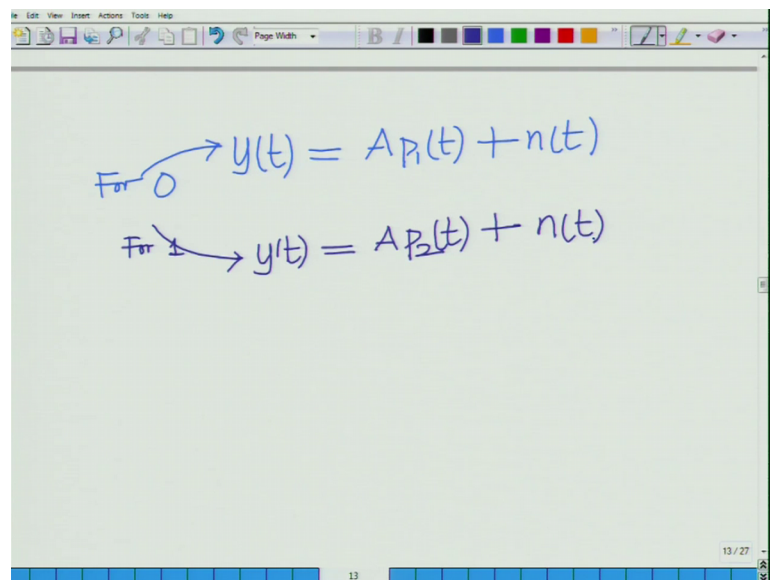
Principles of Communication Systems – Part II
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Lecture - 16

Optimal Decision Rule for FSK, Bit-Error Rate (BER) and Comparison with BPSK, ASK

Hello. Welcome to another module in this massive open online course. So, we are looking at frequency shift keying all right. And in frequency shift keying well we are considering 2 different wave forms, to represent the information bit 0 and one which is shifted in frequency correct.

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For 0 $y(t) = A_P(t) + n(t)$

For 1 $y(t) = A_{P2}(t) + n(t)$

So, we are considering 2 different frequencies f_1 and f_2 since they are wave forms are shifted versions of each other with frequencies this is known as frequency shift key all right. And what we have seen in this is that my $y(t)$ corresponding to 0 is a times $P_1(t)$ plus $n(t)$. That is, this is for information bit 0 for 1, we have $y(t)$ equals a times $P_2(t)$ plus $n(t)$.

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For 1 $\rightarrow y(t) = A p_1(t) + n(t)$

Optimal Receive Filter $h(t)$

$$h(t) = p_1(T-t) - p_2(T-t)$$

Maximizes output SNR.

We have seen the matched filter optimal received filter, optimal receive filter $h(t)$ equals $p_1(T-t) - p_2(T-t)$. So, this is the optimal receive filter, optimal receive filter in the sense we must be well familiar by now optimal in the sense this maximizes the output SNR or signal to noise power ratio.

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$h(t) = p_1(T-t) - p_2(T-t)$

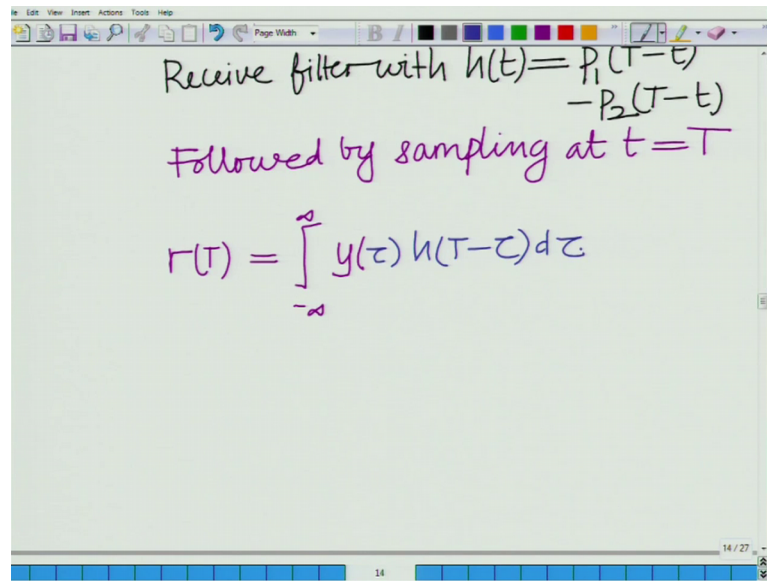
Maximizes output SNR

Consider information bit 0.

$$y(t) = A p_0(t) + n(t).$$

Now, consider the transmission of information bit 0, consider or consider information bit 0. We have for this $y(t)$ equals A times $p_0(t)$ plus $n(t)$.

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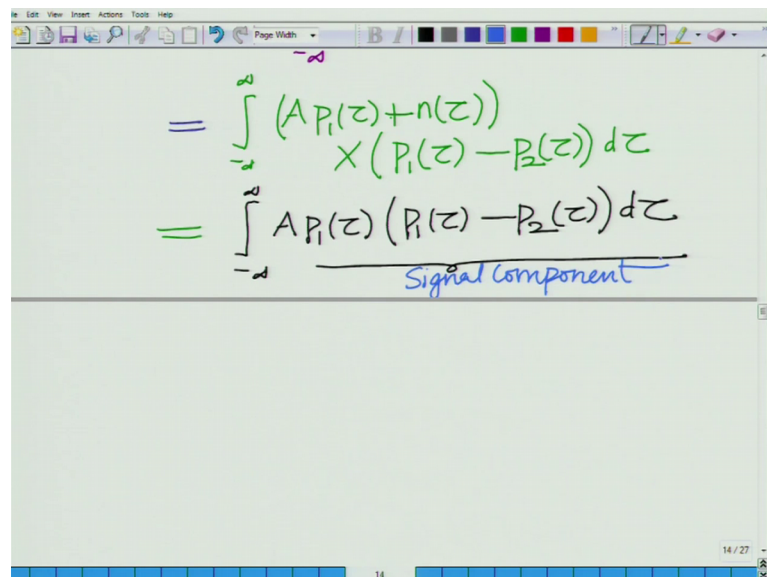


Receive filter with $h(t) = P_1(T-t) - P_2(T-t)$
 Followed by sampling at $t = T$

$$r(T) = \int_{-\infty}^{\infty} y(\tau) h(T-\tau) d\tau$$

Now, matched filter or receive filter with $h(t)$, $h(t)$ equals $P_1(T-t) - P_2(T-t)$ followed by sampling at $t = T$. Now that gives us if you can look at it, we have $r(T) = \int_{-\infty}^{\infty} y(\tau) h(T-\tau) d\tau$ substituting for $h(t)$.

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$$= \int_{-\infty}^{\infty} (A P_1(\tau) + n(\tau)) \times (P_1(\tau) - P_2(\tau)) d\tau$$

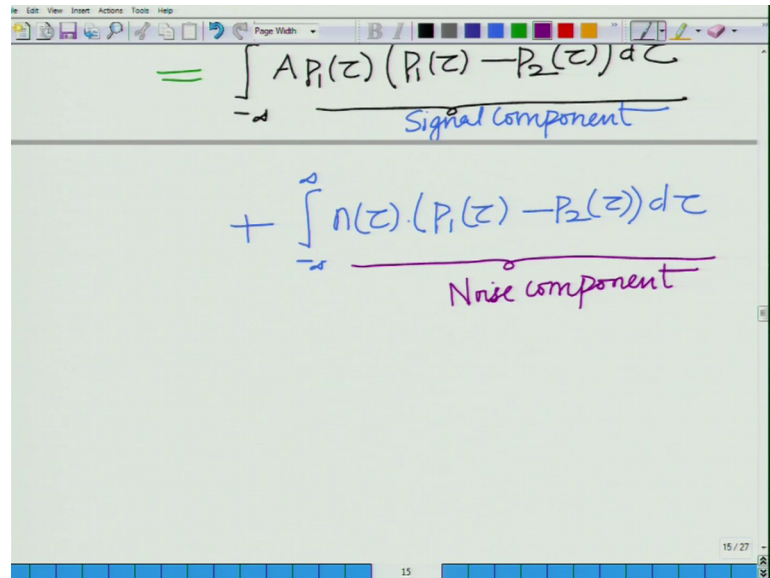
$$= \int_{-\infty}^{\infty} A P_1(\tau) (P_1(\tau) - P_2(\tau)) d\tau$$

Signal Component

This gives us integral this gives us integral minus infinity to infinity $y(t)$ is basically you are well this is a P_1 times P_1 of τ a P_1 of τ , of course, there is noise also a P_1 of τ plus $n(\tau)$ times now the filter $h(t)$ capital H capital T minus τ P_1 which is you

can see is simply $P_1(t) - P_2(t)$, and that gives us well that can be split naturally into 2 terms, $A P_1(t)$ into $P_1(t) - P_2(t)$. This is your signal component, plus integral minus infinity to infinity $n(t)$, $P_1(t) - P_2(t)$ and this is your noise component.

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The image shows a digital whiteboard with a toolbar at the top. The handwritten text is as follows:

$$= \int_{-\infty}^{\infty} A P_1(\tau) (P_1(\tau) - P_2(\tau)) d\tau$$

Signal component

$$+ \int_{-\infty}^{\infty} n(\tau) (P_1(\tau) - P_2(\tau)) d\tau$$

Noise component

The whiteboard interface includes a menu bar (Edit, View, Insert, Actions, Tools, Help), a toolbar with various drawing tools, and a status bar at the bottom showing '15 / 27'.

So, corresponding to the transmission of information signal the inform corresponding to information bit 0, we have computed what is the output after filtering with $h(t)$ followed by sampling at t equal to capital T where capital T is the symbol duration these are in the signal component and the noise component.

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Handwritten derivation on a digital whiteboard showing the simplification of the signal component. The first line is $\text{Signal} = \int_{-\infty}^{\infty} A p_1(\tau)(p_1(\tau) - p_2(\tau)) d\tau$, with a bracket above the integrand labeled "Noise component". The second line is $= A \int_{-\infty}^{\infty} p_1^2(\tau) d\tau - A \int_{-\infty}^{\infty} p_1(\tau)p_2(\tau) d\tau$. The first integral is labeled $E_p = 1$ and the second is labeled "inner product of $p_1(\tau), p_2(\tau)$ = 0".

Now, let us simplify both these components individually now let us come to the signal component. Now the signal equals minus infinity to infinity a times P 1 tau into P 1 tau minus P 2 tau d tau, which is equal to integral minus infinity to infinity a times P 1 square tau d tau minus A times integral minus infinity to infinity P 1 tau P 2 tau d tau. Now you can see this is nothing, but the energy of the pulse this is equal to 1 and this integral minus infinity to infinity P 1 tau P 2 tau d tau, we have seen this is the inner product of P 1 tau comma P 2 tau which is equal to 0. So, this is my product is 0 because P 1 tau and P 2 tau we have said are orthonormal basis functions of the signal space.

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Handwritten derivation on a digital whiteboard. The first line is $= A \cdot E_p = A$, with an arrow pointing to it labeled "Signal component". The second line is $\tilde{n} = \int_{-\infty}^{\infty} n(\tau)(p_1(\tau) - p_2(\tau)) d\tau$.

Therefore all that is left is a times E_p substituting E_p equal to 1, this is remember the signal component after matched filtering. Now if you look at the noise component \tilde{n} , which is equal to minus infinity to infinity $n(\tau) P_1(\tau) - P_2(\tau) d\tau$.

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The image shows a digital whiteboard with the following handwritten content:

$$\tilde{n} = \int_{-\infty}^{\infty} n(\tau)(P_1(\tau) - P_2(\tau))d\tau$$

An arrow points from the text "Gaussian, mean = 0." to the variable \tilde{n} in the equation above.

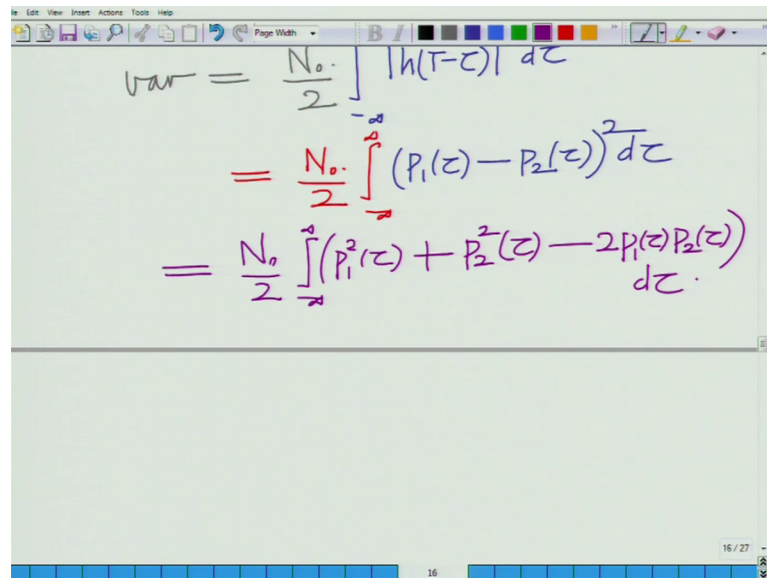
$$\text{var} = \frac{N_0}{2} \int_{-\infty}^{\infty} |h(\tau)|^2 d\tau$$

$$= \frac{N_0}{2}$$

The whiteboard interface includes a menu bar (Edit, View, Insert, Actions, Tools, Help), a toolbar with various drawing tools, and a status bar at the bottom showing "16 / 27" and a page number "16".

Now, this noise we have previously see in from an analysis of the noise output at the noise at the output of a matched filter we are said the this is the input noise is a Gaussian noise process the output noise process also Gaussian once you sample it you get a Gaussian random variable, if the input noise process is 0, mean the output noise process is also 0 mean. So, \tilde{n} is Gaussian mean equal to 0. Now the variance is $N_0/2$ and this also we have seen integral minus infinity to infinity magnitude $h(\tau)$ square $d\tau$ or $|h(\tau)|^2 d\tau$, which is equal to $N_0/2$, integral minus infinity to infinity well $P_1(\tau)$ or.

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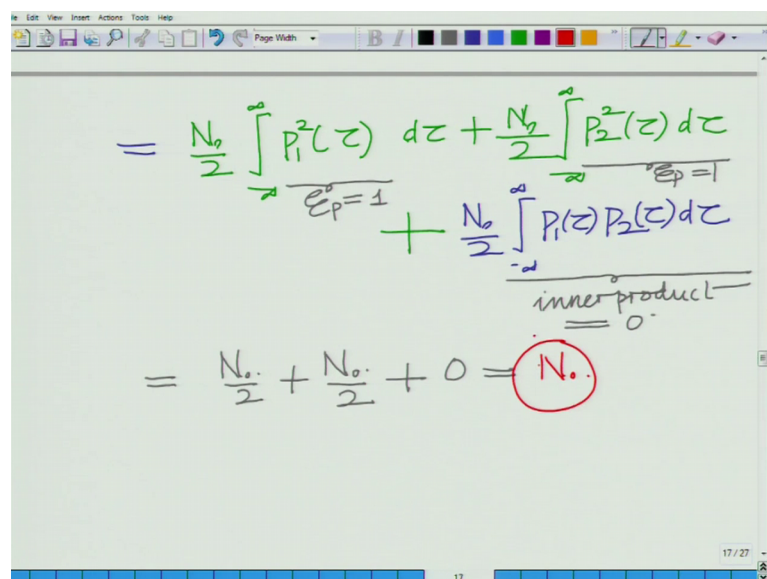


A whiteboard with a presentation software interface at the top. The interface includes a menu bar (File, Edit, View, Insert, Actions, Tools, Help), a toolbar with various icons, and a status bar at the bottom showing 'Page Width' and '16 / 27'. The whiteboard contains the following handwritten mathematical derivation:

$$\begin{aligned} \text{var} &= \frac{N_0}{2} \int_{-\infty}^{\infty} |h(\tau - \tau)| d\tau \\ &= \frac{N_0}{2} \int_{-\infty}^{\infty} (P_1(\tau) - P_2(\tau))^2 d\tau \\ &= \frac{N_0}{2} \int_{-\infty}^{\infty} (P_1^2(\tau) + P_2^2(\tau) - 2P_1(\tau)P_2(\tau)) d\tau \end{aligned}$$

Let me write it for the sake of just being clear let me write this as T minus P 1 tau minus P 2 tau square, d tau. Now this is now expand this P 1 tau minus P 2 tau whole square, what you have is P 1 square tau, plus P 2 square tau minus, twice P 1 tau P 2 tau d tau.

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A whiteboard with a presentation software interface at the top. The interface includes a menu bar (File, Edit, View, Insert, Actions, Tools, Help), a toolbar with various icons, and a status bar at the bottom showing 'Page Width' and '17 / 27'. The whiteboard contains the following handwritten mathematical derivation:

$$\begin{aligned} &= \frac{N_0}{2} \int_{-\infty}^{\infty} P_1^2(\tau) d\tau + \frac{N_0}{2} \int_{-\infty}^{\infty} P_2^2(\tau) d\tau \\ &\quad + \frac{N_0}{2} \int_{-\infty}^{\infty} P_1(\tau)P_2(\tau) d\tau \\ &\quad \text{inner product} \\ &\quad = 0 \\ &= \frac{N_0}{2} + \frac{N_0}{2} + 0 = N_0 \end{aligned}$$

Now, is split this into 2 terms or split this into other 3 terms. This is equal to n naught by 2, integral minus infinity to infinity, P 1 square T minus tau d tau plus n naught by 2 integral minus infinity to infinity P 2 square not T minus tau.

$P_2^2 \tau d\tau$ plus the third term, which is $n \text{ naught by } 2 \int_{-\infty}^{\infty} P_1 \tau P_2 \tau d\tau$. And now you can see once again this is $\int_{-\infty}^{\infty} P_1^2 \tau d\tau$ that is E_p the energy of the pulse which is 1, $\int_{-\infty}^{\infty} P_2^2 \tau d\tau$ this is also E_p which is equal to 1. This is once again this is inner product between P_1 and P_2 , which is equal to 0. So, what is remaining is basically if you can see the noise power or the noise variance that is $n \text{ naught by } 2$, plus $n \text{ naught by } 2$ we are setting E_p equal to 1 plus 0, which is equal to $n \text{ naught}$ $n \text{ naught by } 2$ plus $n \text{ naught}$ which is equal to 0 this is a noise power at the output after sampling.

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$$\begin{aligned}
 & E_p = 1 + \frac{N_0}{2} \int_{-\infty}^{\infty} P_1(z) P_2(z) dz \\
 & \quad \text{inner product} \\
 & \quad = 0 \\
 & = \frac{N_0}{2} + \frac{N_0}{2} + 0 = N_0
 \end{aligned}$$

Noise power after sampling.

So, this is the noise power. Previously we had always noise power $n \text{ naught by } 2$.

But now we have noise power after sampling equals after sampling equals $n \text{ naught}$. So, observe that this is basically once again slightly different with respect to what we had for amplitude shift keying and also binary phase shift keying as noise power, after match filtering and sampling was $n \text{ naught by } 2$ now we have noise power $n \text{ naught}$.

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Handwritten notes on a whiteboard:

$$r(T) = A\epsilon_p + \tilde{n}$$

$$= A + \tilde{n}$$

Gaussian
mean = 0
var = N_0

Corresponding to bit 1

$$y(t) = A p_2(t) + n(t).$$

So, basically the output is $r(T)$ equals well $A \epsilon_p$ plus \tilde{n} setting ϵ_p equal to 1 this is a times writing ϵ_p again because it is. So, that the expression is general and can be valid also for scenarios where ϵ_p is not necessarily equal to 1 \tilde{n} this is Gaussian with mean equal to 0, and variance is equal to N_0 . Now similarly corresponding to the transmission of information bit corresponding to bit 1, well corresponding to bit one we have $y(t)$ equals $A p_2(t)$ plus $n(t)$.

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Handwritten notes on a whiteboard:

Matched Filter

$$h(t) = p_1(T-t) - p_2(T-t)$$

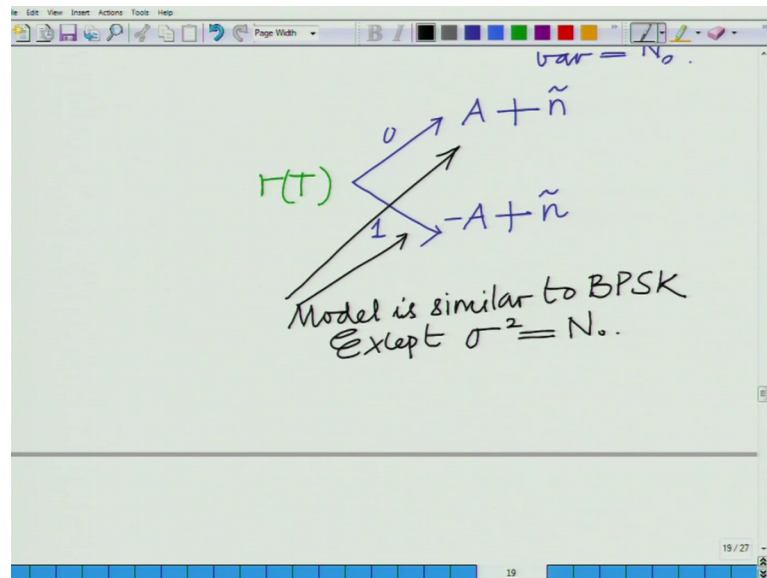
$$r(T) = -A\epsilon_p + \tilde{n}$$

$$= -A + \tilde{n}$$

Gaussian
mean = 0
var = N_0

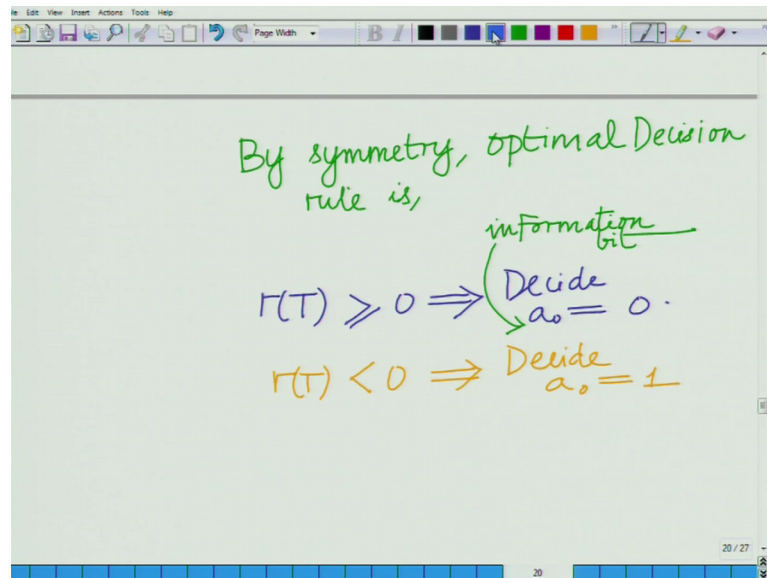
Now, when we matched filter this, or receive filter this, with once again $h(t)$ equals $P_1(t - T) - P_2(t - T)$ the output will be well, you can verify this $r(t)$ is minus A plus \tilde{n} again setting equal to 1, this is equal to minus A plus \tilde{n} or minus simply a minus A plus \tilde{n} , where \tilde{n} is Gaussian. Once again is a same thing the mean equal 0 the variance is equal to N_0 .

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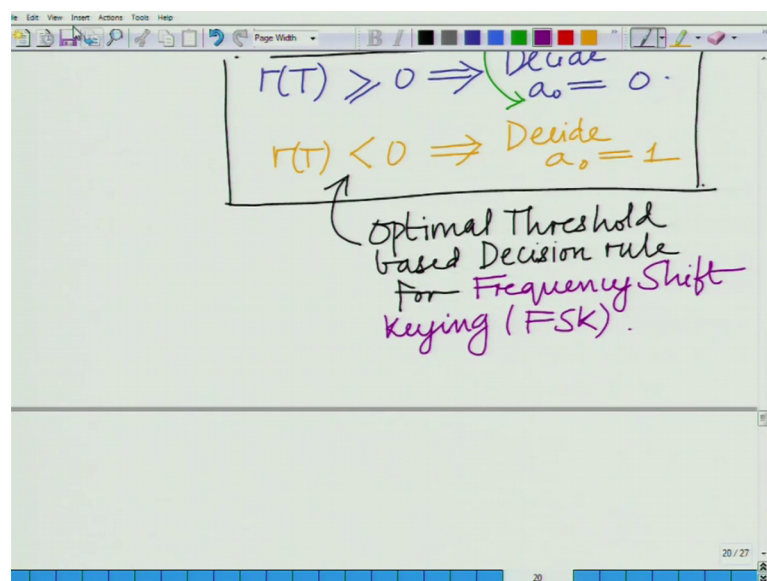
Therefore, now if you can look, if you look at this team comprehensively I can summarise this as the output after filtering and sampling is basically given as corresponding to 0 and 1, this is A plus \tilde{n} this is minus A plus \tilde{n} . Now you can see this is similar to BPSK this model, if you can observe closely, model is similar to BPSK that is binary phase shift keying except noise variance is sigma square equals N_0 , N_0 by 2 right. In BPSK in the noise variance what we are computed \tilde{n} that is noise variance after filtering and sampling was N_0 by 2, here it is N_0 .

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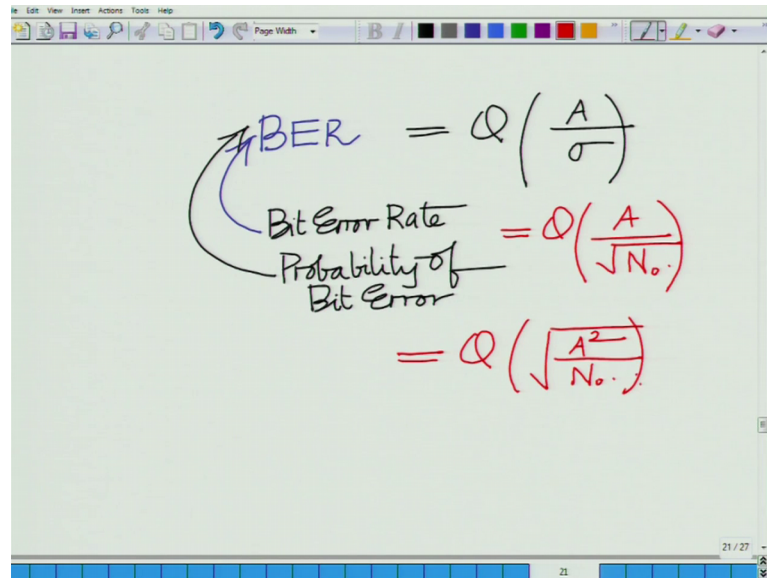
Therefore, now the threshold again by symmetry, look at it the mean is shifted for 0 the mean shifted to a for 1, the mean is shifted to minus A, symmetry by symmetry again the optimal detection rule is to compare the sample r_T with 0. So, once again by symmetry by symmetry, the optimal decision rule, optimal decision rule is the optimal decision rule is once again that is your r_T greater than equal to 0, implies decide well decide a naught equal 0 that is this is the information bit. On the other hand if r_T is less than 0 that implies decide a naught is equal to 1. So, therefore, that summarises your decision rule.

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This is the optimal decision of FSK is not very different from that. In fact, very similar to that of b s k this is the optimal. In fact, we can say threshold based decision rule for frequency shift keying FSK or frequency shift keying, that is f frequency shift keying FSK.

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The image shows a handwritten slide with the following equations and text:

$$BER = Q\left(\frac{A}{\sigma}\right)$$

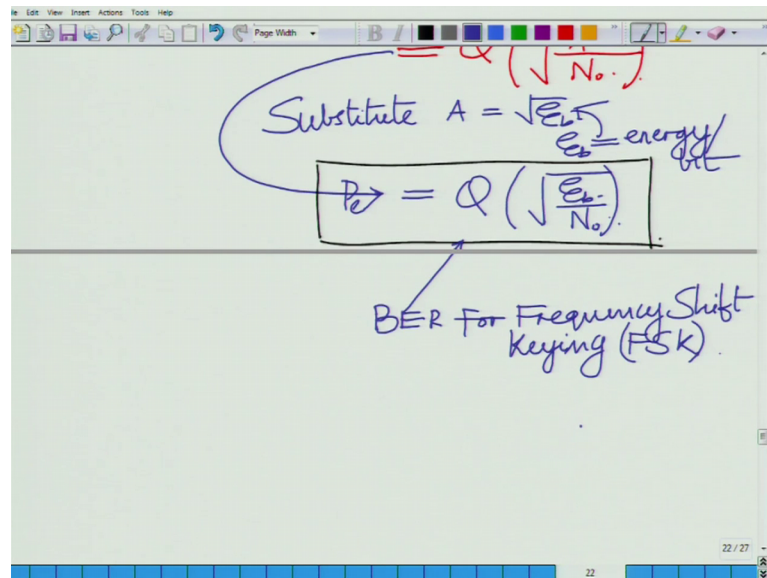
Bit Error Rate = $Q\left(\frac{A}{\sqrt{N_0}}\right)$

Probability of Bit Error = $Q\left(\sqrt{\frac{A^2}{N_0}}\right)$

A blue arrow points from the text "Bit Error Rate" to the first equation. A red arrow points from the text "Probability of Bit Error" to the second equation.

And the corresponding bit error rate again we can see, corresponding bit error rate there is if you look at this bit error rate, which is either which is again another thing that you should be familiar very familiar with, by now this is the bit error rate bit error probability or this known by many terms or the probability, this is equal to Q times A divided by sigma which is equal to Q times A divided by square root of n naught, which is equal to Q times A square divided by n naught.

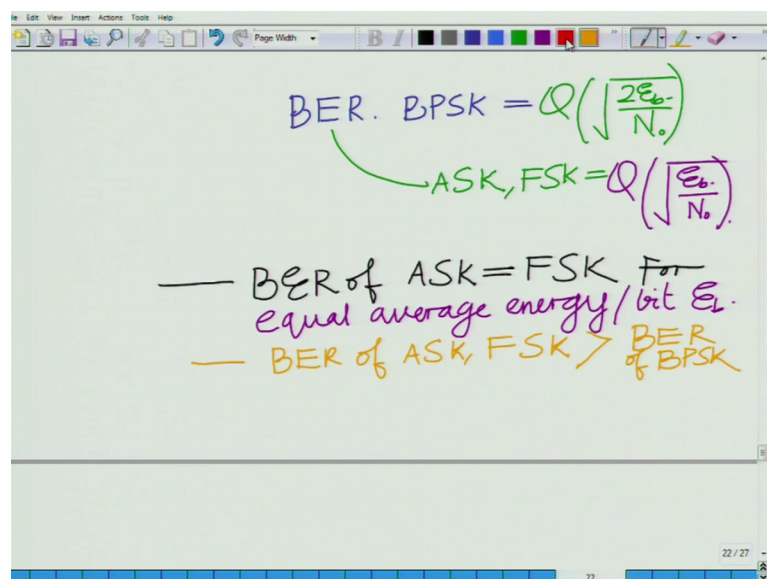
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A screenshot of a presentation slide showing handwritten notes. At the top, there is a red equation $= Q(\sqrt{N_0})$. Below it, the text "Substitute $A = \sqrt{E_b}$ " is written, with a note " $E_b = \text{energy/bit}$ " next to it. A box contains the equation $P_b = Q(\sqrt{\frac{E_b}{N_0}})$. An arrow points from this box to the text "BER for Frequency Shift Keying (FSK)".

Now, substitute A equal to E_b substitute A equal to square root of E_b , as where E_b as we have seen before equals the energy per bit. So, bit error this becomes Q square root of E_b over N_0 this you can say is the probability of bit error or bit error rate. So, this is the bit error rate for frequency shift keying, let us note that. This is the bit error rate this is the bit error rate for frequency shift keying that is FSK. So, this is the bit error rate for frequency shift keying, and again you can see if you compare it with bit error rate now bit error rate for.

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A screenshot of a presentation slide showing handwritten notes. At the top, the equation $BER_{BPSK} = Q(\sqrt{\frac{2E_b}{N_0}})$ is written in green. Below it, an arrow points to the equation $ASK, FSK = Q(\sqrt{\frac{E_b}{N_0}})$ also in green. Below these, two lines of text are written: "— BER of ASK = FSK for equal average energy/bit E_b " and "— BER of ASK, FSK > BER of BPSK".

Now, if you look at bit error rate for BPSK that is basically $Q(\sqrt{2 E_b / N_0})$. And bit error rate for both amplitude shift keying comma frequency shift keying is $Q(\sqrt{E_b / N_0})$. And if you remember or discussion our previous discussion the Q function is the yield probability of a Gaussian random of the standard Gaussian random variable therefore, the Q function is decreasing in it is argument and therefore, this implies that the bit error rate of both frequency shift keying and amplitude shift keying for an average bit energy of E_b is lower the bit error, I am sorry bit error rate is higher the bit error rate performance is worst because remember the Q function is decreasing therefore, $Q(\sqrt{2 E_b / N_0})$ is less than $Q(\sqrt{E_b / N_0})$.

So, the bit error rate of both amplitude shift keying and frequency shift keying is higher and how much and how in efficient our amplitude shift keying and frequency shift keying in comparison to binary phase shift keying we have seen previously that they are 3 dB worse in comparison to binary phase shift keying the reason being you need twice the average bit energy that is you need twice the average bit energy E_b to achieve the same bit error rate performance in amplitude shift keying and frequency shift keying as that of BPSK. So, let us also note that once again all though that should be plenty clear. So, the first point is BER of ASK is identical to that of FSK; obviously, the comparison has to be fair for equal for equal average energy per bit E_b . Further the BER of ASK comma FSK is greater than BER of BPSK.

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equal average energy

— BER of ASK, FSK $>$ BER of BPSK
Because $Q(\cdot)$ Function is a decreasing Function

— ASK, FSK are 3dB inefficient in comparison to BPSK

For same BER, ASK/FSK need 3dB more power.

This is because remember Q function is decreasing in its argument. Gaussian tail probability is a decreasing a monotonically decreasing function. And more importantly we also have the fact that how efficient are there ASK comma FSK amplitude shift keying and frequency shift keying are 3 dB inefficient or 3 dB worse. Which basically means for same bit error rate to achieve the same bit error rate you are ASK amplitude shift keying slash frequency shift keying need 3 dB they need 3 dB more power.

So, they are basically 3 dB to 3 dB worse amplitude shift keying and frequency shift keying at 3 dB worse than binary phase shift keying. So, basically you can say binary phase shift keying for the same average energy per bit E_b achieves the lowest bit error rate amongst BPSK, ASK and frequency shift keying all right. So, BPSK is the amongst these 3 BPSK binary phase shift keying is the most efficient modulation scheme and the reason for that if you explore it will be because binary phase shift keying for the same average bit error rate uses antipodal signalling that is it uses signals plus A and minus A all right which maximises the distance between the constellation points.

You can note this in formally all though we have not shown this regrettably it maximises the distance between the constellation points for the same average bit error rate. While both amplitude shift keying and frequency shift keying have do not do this all right. So, they have a poor performance. So, their performance is poor in comparison to the in comparison to binary phase shift keying which has which we have shown to be in a very efficient digital modulation scheme. So, that completes our discussion on frequency shift keying based on the based on this framework remember of the signal space the concept of signal space where we are now implying not one, but 2 pulse wave forms which are both of unit energy normalised to unit energy all right and also orthogonal to each other.

So, they constitute an orthonormal basis for this signal space we are representing the information bit 0 using a times $\phi_1(t)$ that is along the signal $\phi_1(t)$ basis function $\phi_1(t)$, we are representing the information bit one by a times $\phi_2(t)$ which is along the direction in the signal space or which is along the basis function $\phi_2(t)$ and therefore, we are now using a 2 dimensional signal space in frequency shift keying that is the interesting aspect about frequency shift keying all right.

So, we will stop here.