

Fundamentals of Acoustics
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Lecture – 59
Discrete Fourier Transform

Hello, welcome to Fundamentals of Acoustics. Today is the 5th day of the 10th week of this course, yesterday we had started our discussion on a discrete fourier transform and we had shared with you a relation which will help you transform the discrete data in time domain into frequency domain and the expression for that was showed here.

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NOT USABLE IN MOST EXPERIMENTS

DISCRETE FOURIER TRANSFORM

$$X_k = \sum_{n=0}^{N-1} x_n e^{-i \frac{2\pi k n}{N}}$$

(FORWARD)
TIME DOMAIN \rightarrow FREQ. DOMAIN

x_n : N data points in time domain. } RAW INPUT DATA.
 $x_0 \ x_1 \ x_2 \ x_3 \ \dots \ x_{N-1}$

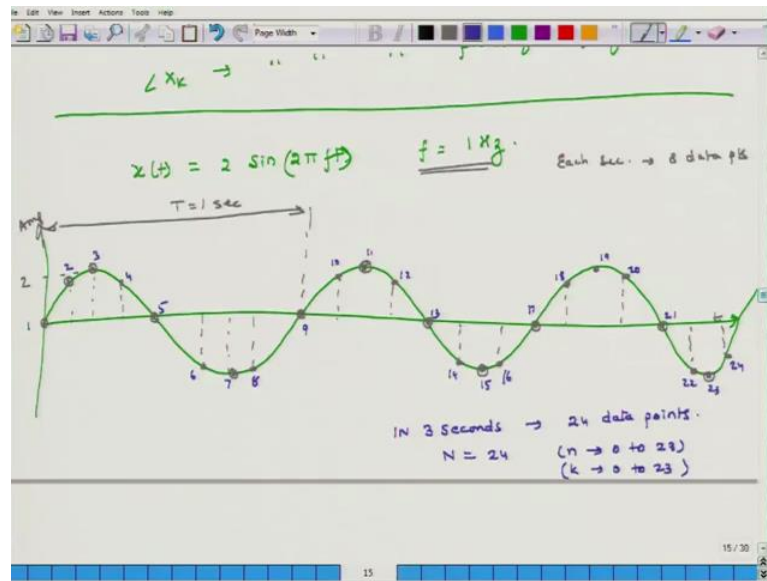
n : Index for time.
 $n=0 : t_0 \quad n=1 : t_1 \quad n=2 : t_2 \quad \dots \quad n=N-1 : t_{N-1}$

X_k : Amplitude of signal at k^{th} frequency.
 $k=0$ 1st frag. $k=1$ 2nd frag. \dots $k=N-1$ Nth frag.

k : Index in frequencies.
Time \rightarrow N frequency

So, that is the expression and we had explained the importance of parameters N, lower case x k and upper case K.

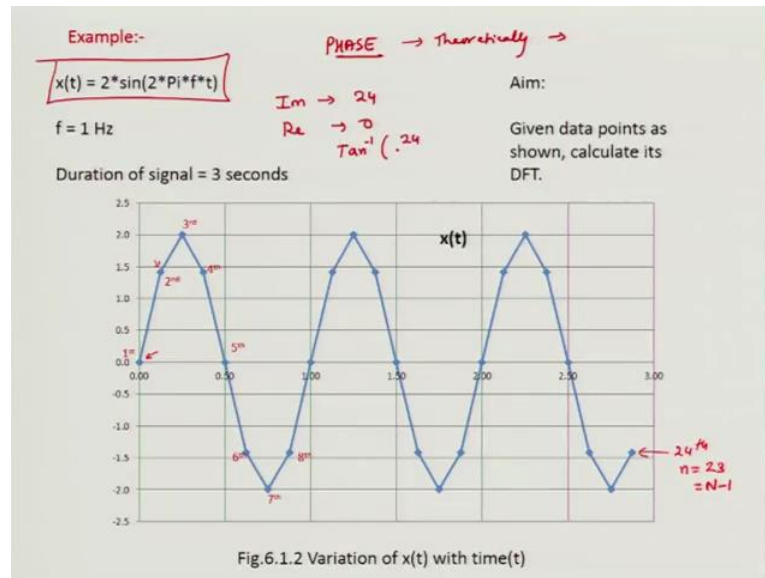
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And now what we are going to do is we have a signal and let say that the signal is X equals 2 times sin of 2 pi f t and let us set f has 1 hertz.

So, the actual signal is shown in green continuous line, but modern data individual systems do not acquire continuous data because they are digital. So, instead of continuous signal, what we are data acquisition system is just getting points and let say we have recording the first 24 points that is for a period of 3 seconds, so I have these 24 points which I have marked here. Now what I want to do is, I want to use this information for 24 points and from that information I want to figure out the Fourier components of this signal that is it is representation in the frequency domain.

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So, with that let us look at some of the slides. So, these are the points, this is point number 1 which corresponds to n is equal to 0. So, this is point number 1 corresponding to n is equal to 0, this point corresponds to n is equal to 2 and so on and so forth and this is the 24th point and the corresponding value of n is 23, which is equal to capital N minus 1. So, this is the actual data which we are getting and now from these data we have to figure out the frequency domain representation of the signal. So, what we do?

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Discrete Fourier Transform

- As seen in Figure 6.1.2, there are eight points measured each second. Hence, our *sampling frequency* f_s is 8.
- Both, n , and k , assume 24 different values, i.e. from 0, to 23. The number "24" corresponds to total number of points present in our data set $x(t)$.
- X_k can assume 24 different values, i.e. X_0 to X_{23} . Further, X_k need not be a real number. It can be complex as well, as the R.H.S. of Eq. 6.1.2 has real as well as imaginary components. Hence, we can fill up the following table.

n	t	$f(t)$	$X_{k=0}$	$X_{k=0}$	$X_{k=23}$	$X_{k=23}$
0								
1								
.								
.								
23								
Sum			$X_{r,0}$	$X_{i,0}$	$X_{r,23}$	$X_{i,23}$

So, first let us look at this table, we have explained that there are 8 points measured each second; our data acquisition system the way on which we are using it is measuring 8 points every second. So, we say that our sampling frequency f_s is 8, 8 samples per second, if it was measuring 16 samples then it would have been 16 and so and so forth. The other thing to notice that both n ; n is what it is the induction time and k which is the induction on frequency, they can assume 24 different values.

So, in this table n is going down and it starting from 0 1 2 3 4 till 23, and k is starts from here and it is going in this direction. So, k is equal to 0 here and I had said that when we compute X_k it will have a real part and an imaginary part. So, for each value of k I have X_r which is the real part and X_i which is the imaginary part. So, I have 2 columns for each value of k , one is this column is the real part, this column is the imaginary part for k is equal to 0 then for k is equal to 1, I will again have a real part and an imaginary part.

So, in this table I have constructed this table which have been used to compute different values of X_k is specifically there different values of X_k their real and imaginary parts and we learn how to do that. So, that is what it is says here, X_k can assume 24 different values from k is equal to 0 to k is equal to 23 and further X_k need to be a real number it can be complex as the RHS of the equation, which we had discussed a real it can have real and imaginary components.

So, what we have in this table is a matrix and here onwards we are going to compute different values of X_k and n goes down and it goes down from 0 to 23, the next column is time, the third column is $f t$. So, this is what there is now we learn how to do it.

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Table 6.1.1

Given $2 \sin(2\pi f t)$ $f=1$

$X_{k=0}$ X_1 X_2 X_3 X_4

k 's

n	t	f(t)	0	0	1	1	2	2	3	3	4	4
0	0.000	0	0	0	0	0	0	0	0	0	0	0
1	0.125	1.414	1.414	0.0000	1.366	-0.3660	1.225	-0.7071	1.000	-1.0000	0.707	-1.2247
2	0.250	2.000	2.000	0.0000	1.732	-1.0000	1.000	-1.7321	0.000	-2.0000	-1.000	-1.7321
3	0.375	1.414	1.414	0.0000	1.000	-1.0000	0.000	-1.4142	-1.000	-1.0000	-1.414	0.0000
4	0.500	0.000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000
5	0.625	-1.414	-1.414	0.0000	-0.366	1.3660	1.225	0.7071	1.000	-1.0000	-0.707	-1.2247
6	0.750	-2.000	-2.000	0.0000	0.000	2.0000	2.000	0.0000	0.000	-2.0000	2.000	0.0000
7	0.875	-1.414	-1.414	0.0000	0.366	1.3660	1.225	-0.7071	1.000	-1.0000	0.707	1.2247
8	1.000	0.000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000
9	1.125	1.414	1.414	0.0000	-1.000	-1.0000	0.000	1.4142	1.000	-1.0000	-1.414	0.0000
10	1.250	2.000	2.000	0.0000	-1.732	-1.0000	1.000	1.7321	0.000	2.0000	1.000	1.7321
11	1.375	1.414	1.414	0.0000	-1.366	-0.3660	1.225	0.7071	-1.000	-1.0000	0.707	1.2247
12	1.500	0.000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000
13	1.625	-1.414	-1.414	0.0000	1.366	-0.3660	-1.225	0.7071	1.000	-1.0000	-0.707	1.2247
14	1.750	-2.000	-2.000	0.0000	1.732	-1.0000	-1.000	1.7321	0.000	-2.0000	1.000	1.7321
15	1.875	-1.414	-1.414	0.0000	1.000	-1.0000	0.000	1.4142	-1.000	-1.0000	1.414	0.0000
16	2.000	0.000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000
17	2.125	1.414	1.414	0.0000	-0.366	1.3660	-1.225	-0.7071	1.000	-1.0000	0.707	1.2247
18	2.250	2.000	2.000	0.0000	0.000	2.0000	2.000	0.0000	0.000	2.0000	2.000	0.0000
19	2.375	1.414	1.414	0.0000	0.366	1.3660	-1.225	0.7071	-1.000	-1.0000	0.707	-1.2247
20	2.500	0.000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000	0.000	0.0000
21	2.625	-1.414	-1.414	0.0000	-1.000	-1.0000	0.000	-1.4142	1.000	-1.0000	-1.414	0.0000
22	2.750	-2.000	-2.000	0.0000	-1.732	-1.0000	-1.000	-1.7321	0.000	-2.0000	1.000	-1.7321
23	2.875	-1.414	-1.414	0.0000	-1.366	-0.3660	-1.225	-0.7071	-1.000	-1.0000	-0.707	-1.2247
FFT (sum)			0	0	0	-7E-16	-4E-15	-3E-15	1E-14	-24	3E-15	5E-15
Mag of FFT			0	0	7E-16	5E-15	24	6E-15				

So, this is what we do. So, this is an illustration how for each value of k we can compute X_k . So, first n is equal to 0 to 23 it has been listed here, then at a given value of time, when n is equal to 0, time is 0, when n is equal to 1 what is the value of time we are measuring 8 points each second right. What does that mean? It means that each second is divided into 8 different parts, which means that corresponding to n is equal to 1, this is 1 over 8th, t is 1 over 8th which is 0.125 seconds, for n is equal to 2 it is 0.25 seconds and so on and so forth. So, the final value of t corresponding to n is equal to 23 is 2.85 seconds.

Now this formula was already given and what was this formula. So, for each time I have to know what is its value. So, what was the value? It was $2 \sin 2 \pi f t$ and f was 1 hertz. So, for this value of time I have calculated the value of function it is given here. So, all these data is given from this given data which we can measure experimentally we have to compute capital X right X_k is X_1, X_2, X_3, X_4 , that is what is shown here.

So, these 2 columns correspond to k is equal to 0, these 2 columns correspond to actually it should not k, but X_k this column corresponds to X_1 , this corresponds to X_2 , this corresponds to X_3 and so on and so forth. So, what do we do? So, we will explain that in the next slide. So, let say we want to compute X_0 , we already know the values of f_0, f_1, f_2, f_3 , from this column.

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Discrete Fourier Transform

- Let us have some data points between $x_n(t)$ and t . Then this can be easily converted to DFT form using,

$$X_k = \sum_0^{N-1} x_n(t) \left[\cos\left(\frac{2\pi kn}{N}\right) - i \sin\left(\frac{2\pi kn}{N}\right) \right] \quad \text{Eq.6.2.1}$$

When $k=0$;

$$X_0 = \sum_0^{N-1} x_n(t) \left[\cos\left(\frac{2\pi \cdot 0 \cdot n}{N}\right) - i \sin\left(\frac{2\pi \cdot 0 \cdot n}{N}\right) \right]$$

When $k=1$;

$$X_1 = \sum_0^{N-1} x_n(t) \left[\cos\left(\frac{2\pi \cdot 1 \cdot n}{N}\right) - i \sin\left(\frac{2\pi \cdot 1 \cdot n}{N}\right) \right]$$

Handwritten notes in the image:
 - A red box highlights the general equation (Eq.6.2.1).
 - A red arrow points from $e^{-i\omega kn}$ to the bracketed term in the equation.
 - A red arrow points from "Eq.6.2.1" to the equation.
 - A red circle highlights X_0 in the $k=0$ case.
 - A red arrow points from "summation" to the summation symbol in the $k=1$ case.
 - Red arrows point from the 1 in the $k=1$ case to the 1 in the cosine and sine terms.

So, what is X_0 ? This is the general relation, X_k equals X_n which actually changes with time, times what is this thing? This is e to the power of minus $i\omega k$ what is the k n divided by N . So, I can express it as a cosine term and the sin term, I can express this $e^{i\omega k n}$ as a sin and the cosine term.

So, what I have done is I have expanded this and in this relation I have put k is equal to 0. So, when k is equal to 0 this expression becomes X_0 equals sum of X_n cosine 2π times 0 times n divided by capital N minus i sin 2π times 0 times small n divided by capital N ; how did I get this? I use this equation and then this equation, I have put k equals 0 and then for different values of n I have summed this entire expression for different values of n small n and n changes from 0 to N minus 1. So, in that way I get X_0 . If I have to when calculate X_1 again we go back to this equation and I plug in k equals 1. So, when k is equal to 1 I have 1 here, and using in this expression I do the summation this is the summation and I get X_1 . So, in this way I have been able to calculate X_0 , X_1 , X_2 , X_3 , and so on and so forth. So, these 2 columns are for X_0 , these 2 columns are for X_1 , these 2 columns are for X_2 and so on and so forth and as I said each X will have a real and an imaginary part. So, I have listed those here.

So, what I have done is here this is the individual component corresponding to each n , what is the individual component? So, if N is equal to 1 then when I put n is equal to when in this expression I get the first component right. So, these are the individual

components and this is the sum of the entire column. So, this is the sum of the entire column corresponds to this operation and. So, what and then finally, I get the magnitude of a 50. This magnitude of a 50, this magnitude is basically this term square plus this term square and the square root of 2 is square root of the sum of the squares, this is the real this is the real component, this is the imaginary component, if I square them add them up and take the square root that is the magnitude of X_k .

Similarly, for X_1 square of this plus square of this and then add them up and take the square root I get this is X_1 the last row; this is for X_2 this is the real portion, this is the imaginary portion, this is the square root of sum of their squares so that is X_2 and similarly this is X_3 . So, in X_3 what do we see? This is the real portion, this is the imaginary portion and the sum of them is giving as X_3 and so on and so forth. So, what does it tell us? X_0 equals 0, X_1 equals 7 E minus 16 which is 0, X_2 equals 5 E minus 15 which is 0, X_3 equals 24, X_4 equals 0 and so on and so forth.

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• X_k amplitude is not the actual amplitude related to actual frequency but it can be converted to actual by using a simple mathematical expression as shown below,

$$X_{ka} = \left(\frac{2X_k}{N} \right) \quad \text{Eq. 6.2.2}$$

Where, X_{ka} = Actual amplitude with respect to actual frequency
 N = Total number of data points

• If we plot X_k Vs k then it will be symmetric about $k = (N/2 - 1)$ thus we can consider only 1st half portion.

Handwritten calculations on the slide:

$$X_{0a} = 0 \quad X_{2a} = 0$$

$$X_{1a} = 0 \quad X_{3a} = \frac{2 \times 24}{24} = 2$$

So, what did we get? We got X_0 is equal to 0, X_1 equals 0, X_2 equals 0, X_3 equals 24, X_4 equals 0 and it went on till X_{23} because there are 24 points, what does this mean that the first frequency has 0 amplitude, second frequency has 0 amplitude, third frequency has 0 amplitude, but fourth frequency has an amplitude of 24.

Now, this is not the real amplitude as I mentioned earlier, we have to convert into some real units. So, what is the real amplitude? The relation for real amplitude is given

here. So, that is equal to X_k , k a means actual, so this is actual, this equals 2 times X_k this is the relation 2 times X_k divided by N . N is.

Student: Number of points.

Number of points; which mean that X_0 a equals 0, X_0 no; X_1 a equals 0, X_2 a equals 0, but X_3 a equals.

Student: 2 (Refer Time: 13:50).

2 into X_k . X_k is 24 divided by number of points 24 is 2, but we still do not know, but this X_3 corresponds to how many hertz. All we know is that this is the fourth frequency in the system, first frequency actual amplitude the 0, second frequency actual amplitude to 0, third frequency actual amplitude to 0, fourth frequency amplitude is 2 that is all we know.

Now, we will figure out how do we compute the frequency of the system. But before we do that this is an important thing to note, it is says if we plot X_k versus k . So, I am plotting X_k here and I am plotting k on the X axis, then at this location k is equal to N over 2 minus 1. So, let say this is k is equal to N over 2 minus 1 this function it will be symmetric. So, if it rises here it will rise here, if it goes down here it will go down here. So, at this location X_k is equal to N by 2 minus 1, the plot of X_k versus k will be symmetric and typically we take data only from the first half or the second half because all that data is symmetric.

So, what we have learnt till so far is 2 things; how do we calculate different values of X_k we have learned that and from those X_k we can compute the actual amplitudes of individual frequencies, but what are those actual frequencies we do not know that. So, in the next few slides we will learn that as well. So, this is the thing for the remaining portion.

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- Nyquist Theorem (Sampling Theorem): ←

According to the Nyquist Theorem, the sampling rate must be at least $2f_{max}$ or twice the highest analog frequency component.

Maximum possible frequency can be extracted from signal = $f_s/2$ ←

Where, f_s is sampling rate (Hz).

→ • Normalized frequency = k/N ←

• Normalized circular frequency = $2\pi k/N$ ←

• Actual frequency (Hz) = $k f_s/N$ → f_{actual}

• Circular Actual frequency (rad/s) = $2\pi k f_s/N$

4 Definitions

$f_s = 8$

f_{actual}

k	0
1	$1 \times 8/24 = 1/3$
2	$2 \times 8/24 \rightarrow 2/3$
3	$3 \times 8/24 \rightarrow 1 Hz$
4	
5	
6	
7	
8	

So, before we talk about frequencies let us talk about this Nyquist theorem. So, this is a theorem developed by mister Nyquist and he says that if there is a signal and let say that signal has lots of frequencies, it may have 1 hertz, 2 hertz, 30 hertz ,40 hertz, 20000 hertz and if I have to capture all the frequencies which it has. So, I let say that the highest possible frequency in that signal is 20000 hertz.

Then I should sample it at a frequency which is twice the maximum frequency in the signal $2 f_{max}$. So, if I think that my signal it may have frequencies as I high as 15000 hertz then my sampling frequency it should be at least, it should be at least 2 times that which is 30000 hertz. So, that is what it means. So, another way to look at is that if I am sampling my signal at frequency at value f_s , f_s is sampling frequency when the maximum possible frequency which I can extract from it is f_s divided by 2, this is there. So, this is an important parameter f_s because when we calculate the actual frequencies we will use this.

Now, in DFT, we had actually 4 definitions of frequencies, first definition is known as normalized frequency. So, how many frequencies, frequency components we get if there are n points.

Student: (Refer Time: 18:00).

We will get n number of frequencies right corresponding to k is equal to 0, k is equal to 1, k is equal to 2 and so on and so forth. Now the normalized so for the kth frequency, normalized frequency is defined as K divided by N. So, it will be 0 by N, 1 by N, 2 by N 3 by N and it will go up to.

Student: (Refer Time: 18:30).

N minus 1 by (Refer Time: 18:32), the second frequency is we do not bother too much about this it does not make may a lot of physical sense; the second frequency definition in DFT is normalized circular frequency. So, if you just multiply this K over N by 2 pi this is your normalized circular frequency and the actual frequency is k times f_s divided by N, it is k times f_s divided by N. So, in our case what is the value of f_s?

Student: 8.

In our case f_s was 8. So, k, 0, 1, 2, 3, 4, 5, 6, 7, 8 and so and so forth; actual frequency let us call this f_{actual}. So, let us compute f_{actual}, when k was 0 f_{actual} means 0 hertz when k was 1 it means 1 times 8 divided by 24 is one-third hertz, when k was 2 times 8 divided by 24 it means two-third hertz. When this was 3 it is 3 times 8 divided by 24 it means 1 hertz.

So, the moment we know k using sampling frequency and the total number of points which we have collected, we can calculate actual frequencies and we have already learnt how to calculate the actual amplitude using this relation. So, we know now how to calculate actual amplitude and we found that actual amplitude for the fourth frequency was 2 and that corresponds 2 frequency which equals 1 hertz and that is exactly what are original r result should be, because what was the original signal? The original signal was $2 \sin \text{ times } 2 \pi f t$. So, the frequency was 1 hertz which we have calculated and the amplitude was 2 and the last thing is phase, we can also calculate the phase of this signal, now what is the phase of this signal phase theoretically, it is what?

Student: (Refer Time: 21:20).

So, let us look at. If this also the imaginary component was 24, real component was 0. So, tan inverse 24, no actually sorry it is minus 24 if we look at.

(Refer Slide Time: 21:41)

- If converting this time domain signal into frequency domain then its magnitude can be expressed as,

$$X_k = \sum_{n=0}^{N-1} x_n(t) e^{-j\frac{2\pi kn}{N}} \quad \text{Eq.6.1.1}$$

Points about equation 6.1.1 :-

- ❖ For data set of N points long, n varies from 0 to N-1.
- ❖ $i = \sqrt{-1}$.
- ❖ For data set of N points long, k varies from 0 to N-1.

Eq. 6.1.1 can further written as,

$$X_k = \sum_{n=0}^{N-1} x_n(t) e^{-j\frac{2\pi kn}{N}}$$
$$= \sum_{n=0}^{N-1} x_n(t) \left[\cos\left(\frac{2\pi kn}{N}\right) - i \sin\left(\frac{2\pi kn}{N}\right) \right] \quad \text{Eq.6.1.2}$$

Student: Minus 24.

Where is that? This is minus 24; imaginary component is minus 24, real component is 0. So, from this you can compute the phase also. So, now, you know how to compute phase for each frequency component, how to compute it is amplitude? The amplitude can be calculated by first constructing the table and then using this X_k equals $2 X_k$ divided by N and then the corresponding frequency can be calculated by using this relation.

So, I think that all this information is sufficient for today's lecture, we will meet once again tomorrow and we will discuss some other features of DFT so that when you actually acquire data, you know how to correctly acquire data and how to use it effectively. So, with that we close the discussion for today and we will meet once again tomorrow.

Thank you.