

So Fourier transform is a step forward from what we learned in the last lecture. So Fourier series, now we have transform. So what's the difference? And I will explain. But the prerequisite is that you have a good understanding of Fourier series. Again, for those who came in later, and if you feel really lost in Fourier series, stay in the room after the lecture. We chat a little bit. And it sounds to me majority students are on track. And you may not understand everything. But you need to review and do homework. And you check the chapter I wrote. You check Google using multiple ways and in group discussions. Fourier analysis is so important. I don't want you to get lost now. Otherwise, you will not enjoy medical imaging modalities later on. Anyway, so this is our schedule. We are on schedule, so no problem. And the basic idea, let me reduce to some very fundamental basic idea. You have a general function. The function you know is, say, in one-dimensional case,  $f$  of  $t$ , that's a function of  $t$ . You think it's a time-varying function. And you can view function as it is. And you can think of function in a different way. For example, you think a function is just a summation of many impulses. So we know in the linear system theory, continuous function can be expressed as a convolution of the function and the delta, Dirac delta function. So the meaning of that convolution is basically, say, arbitrary function, you cut it into small pieces. You add all these small pieces together, all these right impulses. Put together, that's just your original function. That's one way to represent function. The same function, and I explained to you, in this case, a periodic function, continuous function. Just think of one copy. Other copy is the same thing. This is one copy of a periodic function. This function can be represented as a summation of many sinusoidal functions. These sinusoidal functions can change in frequency, phase, amplitude. And the trick is, really, you'll find the amplitude, the phase, you'll find all these parameters so that the sinusoidal components added together will recover your original function. So just verbally, these ideas are not challenging at all. Just how function you represent in different ways. One way, view the function as many, many pixels impulses, delta functions, a kind of particle view. The other way, you just decompose the function into multiple wave components, as shown here. So what we learned in the previous lecture,

nothing more than this slice, really.  
It just say arbitrary function, you can represent it as a DC component plus a bunch of cosine components and the number of sine components.  
And the components are different in terms of amplitude, in terms of frequency, because  $n$  can go from 1 to a very, very large number. So the larger the  $n$  value is, the higher the frequency will be.  
Sine, cosine, the two components, sine components, cosine components, for the same frequency, you can combine into a single sinusoidal function with a phase factor.  
That's why I say phase difference.  
But if you decompose into sine and cosine parts, you don't need a phase.  
So just a bunch of DC components, cosine, sine components.  
The trick is, how can you find this coefficient? And the formulas are available for you to do the trick. So to find  $a_0$ , you basically do an integral with respect to  $f$  of  $t$ .  
So this is just a known function.  
 $f$  of  $t$  is a known function.  
You do the integral with respect to  $t$ , and you get  $a_0$ . Likewise, you do computation like this.  
So it's just one slice.  
If you got confused last lecture, you just say, OK, here is a formula.  
Heuristically, I want to represent the function  $f$  of  $t$  as a summation of a sinusoidal component in terms of sine, cosine, and also special case constant.  
How can we find these components, these coefficients? These are formulas.  
So if you do not understand what's going on, remember the formula.  
Then I give you arbitrary continuous function  $f$  of  $t$ . You plug into the formula.  
You get real numbers for  $a_0$ ,  $a$  and  $b$ .  
And then with this  $a$ ,  $b$  coefficient, you just plot these sinusoidal waves.  
Then add them together.  
Magically, you recover the original function  $f$  of  $t$ .  
The larger and the higher accuracy, and the capital  $N$  here can go to infinity.  
So you have a very accurate representation.  
So this is summary really what I told you.  
I explained how you derive, how you view the formulas to compute a coefficient.  
I say that is really inner product.  
Geometrically, it's a projection of this infinitely many dimensional vector.  
In this case, it's a function projected onto one of the basis functions, either sine or cosine or constant.  
Because the constant sine, cosine, with arbitrary integer number  $n$ .  
And these functions together form so-called also normal basis, not much different from three dimensional picture.  
You have arbitrary vector.  
You can represent that vector in terms of its  $x$ ,  $y$ ,  $z$  components.  
That's a geometrical picture.

And the geometrical insight is so important.  
 If you got confused at this moment,  
 I really encourage you to understand in the review  
 process.  
 But the very bottom line, you see,  
 this is a way to represent the function in terms  
 of sinusoidal components.  
 And these are formulas to compute a coefficient.  
 If you understand this part, accept as it is.  
 So this is Fourier series in real form.  
 And I said this real form is kind of complicated.  
 You have three terms.  
 And in complex form, we have a more compact representation  
 of the same thing.  
 So complex form, real form, is just essentially equivalent.  
 And with complex form, we have convenience  
 to write a less number of letters.  
 So just the same thing.  
 And the  $c_n$  is a coefficient that we  
 can compute in terms of inner product  
 between x-pronential basis function  $e$  to the power  
 $minus\ 2\ \pi\ int$ .  
 And in my book, I always put  $i$  first.  
 $Minus\ i,\ 2\ \pi,\ int$ , doesn't matter,  
 with the function to be represented.  
 So the Fourier series can be written in the complex form.  
 So far, we have been saying the periodic function  
 has a unit period over the interval  $0,\ 1$ .  
 And for other intervals, like from  $100$  to  $101$ ,  
 the same thing.  
 You just keep repeating the fundamental picture over  $0,\ 1$ .  
 When the period is not a unit, we  
 say that can be easily extended.  
 So this is the Fourier series for arbitrary function  
 with period capital  $T$ . So I make it symmetric.  
 So you think the function from the  $minus\ t$  divided by  $2$   
 go all the way up to positive capital  $T$  divided by  $2$ .  
 So this is a function over there.  
 And we can express this function in terms  
 of Fourier series in complex form.  
 So this is a constant for  $n$ th element.  
 And we do inner product here.  
 And this is the complex harmonic component.  
 And  $2\ \pi$  is still there.  
 $n$  is there.  
 Then you have  $t$  over capital  $T$ . And  $t$  over capital  $T$   
 kind of normalize lowercase  $t$ .  
 And this is the normalization factor.  
 Capital  $T$ ,  $1$  over capital  $T$ . So the range  
 of this quantity is still from  $0$  to  $1$ .  
 But in this case, it is really from  $minus\ 0.5$   
 to positive  $0.5$ .  
 The periodic function, you think is starting point  
 from negative  $0.5$  to  $0.5$ .  
 Or you think the period start from  $0$  to  $1$ .  
 That's just for convenience.  
 Whatever you view it, as long as you  
 have a whole interval of a total length of  $1$ .  
 And then you have a basic picture.  
 And the other adjacent and continuously many intervals,  
 you just repeat the basic function of form.  
 So this is the formula.  
 So easier way.  
 So what we learned can be summarized in real form,

equivalently in complex form.  
Or you don't want to remember too many.  
Then just remember this one.  
So when you make capital T equal to 1,  
so that covers the complex form.  
This is the unit interval case.  
And you can just do some computation,  
convert complex form back to the real form.  
Real form is nice because we see the geometrical meanings  
crystal clear.  
And with complex form, you need some abstract thinking  
in the complex space.  
Anyway, so this is so much kind of review or warm up  
on what we learned to prepare Fourier transform.  
So by now, you may feel, why bother  
we use Fourier transform?  
So we have a function  $f$  of  $t$ .  
And we talked about how to represent  $f$  of  $t$   
in terms of a bunch of sinusoidal components  
or earlier as a series of delta or impulses  
to represent the original signal.  
You have different ways to view the same thing.  
Usually means you have a better understanding.  
You have a larger flexibility to do things smartly.  
Oftentimes, you have a task to achieve.  
Common sense, you want to find the easy way  
to solve the problem.  
You want to use a simple method to solve either simple  
or complicated problems.  
So the Fourier transformation or the delta function,  
all these alternative representations  
give you alternative ways to view the problem.  
And sometimes, complicated problem in one domain  
will look much simpler in another domain.  
So like Roman numeric, if you write down Roman numeric  
to try to do multiplication, that  
will be very, very complicated.  
But if you just write regular numbers  
or you write in binary numbers, then the multiplication  
can be very easy, either for yourself or for computer.  
So the representation is very important.  
So the Fourier series or Fourier transform  
gives us a way to convert a problem into Fourier domain.  
And as we will see later on, the Fourier representation  
makes some difficult problems very easy.  
So that's one thing.  
Another thing, the original function  
can be quite complicated.  
And we decompose it into some simple form,  
just view it as a collection of some very simple form  
of basis functions.  
So we divide the original function into small pieces.  
And we know how to deal with single sinusoidal or impulse  
function, like the convolution operation was derived  
by using the delta impulse.  
You know how a system responds to an impulse function.  
Then an arbitrary function is decomposed  
into a series of impulse functions.  
Then you add system responses to individual impulse functions  
together.  
So we're talking about two rather important principles.  
So you always pray for simple ways to solve a problem.  
And when the problem is sizable and you  
want to divide and conquer or divide the rule,

this is a very important strategy.  
 So Fourier analysis is something like that.  
 And we learned Fourier series.  
 Now we talk about Fourier transform  
 corresponding to our chapter four in the book  
 draft I shared with you.  
 I think when I got time sometime this month,  
 I will just fix a few typos and polish a little bit further.  
 I will update you with the next version of the foundation part.  
 Anyway, today we are going to talk about how can we derive  
 Fourier transform from Fourier series.  
 Then we talk about some fundamental interesting  
 properties of Fourier transform.  
 And then we talk about a high-dimensional expansion.  
 And in the high-dimensional case,  
 I gave you an example to show you  
 the trouble is really worthwhile.  
 Once you convert an arbitrary image function  
 into Fourier transform, you can do certain things,  
 like you can remove noise very easily.  
 So you will get a better understanding  
 of the utility of Fourier analysis.  
 So you showed these slides before.  
 This is a summary of what we learned  
 in the previous lecture.  
 Also starting point, we used these slides as the first step.  
 Then I just keep deriving so-called Fourier transform.  
 So this is what I just explained to you.  
 Arbitrary function is a periodic function.  
 And the fundamental period is over minus capital T divided  
 by 2 to the other positive half.  
 So you define the function over this interval.  
 Then this arbitrary function can be  
 represented as a bunch of sinusoidal functions  
 in a complex form.  
 So you have a very compact  $e^{i \omega t}$ ,  
 lowercase t over capital T. By Euler formula,  
 we know these sinusoidal components eventually  
 can be broken down to sine and the cosine functions,  
 certainly at a different frequency.  
 This n can go from minus infinity to positive infinity.  
 This is a formula under the bottom line  
 is how you can compute coefficient  $c_n$  in terms  
 of inner product.  
 So  $f(t)$  and this general term, complex harmonic component.  
 So this is summary all we learned in the previous  
 lecture, OK?  
 We can do this, OK?  
 This is  $c_n$ .  
 This is how you compute  $c_n$ .  
 OK, let's just insert this right-hand side into the  $c_n$   
 place.  
 What do you get is something like this, right?  
 OK, we just insert  $c_n$  into the Fourier series  
 in the complex form.  
 We got this, OK?  
 So this is  $1$  over capital T. When  
 we talk about a unit period, we don't have this.  
 But when you have arbitrary interval with period T,  
 you need to do this averaging.  
 So here is a kind of normalization.  
 This is averaging over the whole interval.  
 So we move this  $1$  over capital T out.  
 So you see, this is a summation, OK, weighted by  $1$  over T.

And inside this bracket, you have inner product.  
 And this right-hand side, you notice,  
 and you keep computing inner product.  
 So this integral, I call it inner product,  
 because two things really multiply together.  
 You do pointwise matching.  
 You got the partial products, then add them together.  
 Really here is you are integrating them together.  
 So these are many, many inner product indexed by  $n$ .  
 So  $n$  keep going from negative to positive integers, OK?  
 So for many different  $n$ 's, you keep doing this inner product.  
 And for each  $n$ , you really can think  
 you are trying to evaluate the value at a frequency  
 point  $n$  divided by capital  $T$ . So this is a frequency point.  
 And to understand that better, so let  
 me show you this picture.  
 So you think the frequency point  $n$  divided by  $T$ ,  
 $n$  keep changing from negative to positive.  
 So you have many, many values.  
 So each  $n$  give you one point, OK,  
 when  $n$  equal to  $0$  got this point, DC component.  
 Equal to  $1$ , you get this one.  
 Minus  $1$ , you get this one.  
 For arbitrary  $n$ , you get a frequency,  
 this create a frequency point at the green location.  
 $n$  equal to  $n$  divided by  $T$ , OK?  
 So this is a frequency point.  
 And you just think the interval is from minus capital  $T$   
 divided by  $2$  all the way to capital  $T$  divided by  $2$ .  
 And we can think the  $T$  is very large, OK?  
 You can keep writing  $n$  become larger and larger.  
 So this single period will become so wide,  
 cover the whole  $u$ -axis.  
 And the  $u$  really signifies the frequency axis, OK?  
 This is talking about frequency.  
 Originally, we think Fourier components  
 as discretized frequency for periodic function.  
 But if the period of a periodic function becomes so large,  
 so the minus  $T$  over  $2$ , the  $T$  is so large,  
 minus  $T$  over  $2$  will cross minus infinity.  
 Likewise,  $T$  divided by  $2$  will cross positive infinity.  
 So I'm trying to build a bridge from a periodic function  
 to non-periodic function by writing capital  $T$  approaches  
 infinity.  
 So that way, you really just deal  
 with non-periodic function.  
 Non-periodic function is nothing but a periodic function  
 with infinity along periodic.  
 So anything after infinity along, we don't care.  
 So that's a trick.  
 OK, see, this is a comment of the right-hand side  
 inner product at infinity mean a discrete frequency point,  $u$   
 equal to  $n$  divided by  $T$ . And therefore,  
 sufficiently large  $T$  and all integer  $n$ ,  
 the intervals for  $u$  is dense on the whole number axis.  
 This number axis is just the frequency axis.  
 So you're sampling the point very densely.  
 And you really put the sampling point over the whole  $u$  axis  
 when capital  $T$  is very big.  
 So you just sample the whole  $u$  axis.  
 What's the interval?  
 Interval between adjacent sampling point.  
 So this is  $n$  over capital  $T$ ,  $n$  plus  $1$  divided by capital  $T$ .  
 So this increment  $\Delta u$  equal to  $1$  over  $T$ .

This is nice to know.  
 So the whole trick is really summarized on this slide.  
 So now we can write capital T becomes really big.  
 So this increment becomes very small.  
 And these variables can be reworded  
 in terms of continuous variable.  
 In other words,  $n$  divided by  $T$  can be replaced by  $u$ .  
 So this is shown here.  
 OK, just look at this slide carefully.  
 Then we know the essential point of this lecture  
 and how we derive the so-called Fourier transform,  
 continuous transform, how we can derive inverse Fourier  
 transform.  
 So just try to follow me here.  
 So this is what we got.  
 When we insert  $C_n$ , the expression for  $C_n$ ,  
 in the place  $C_n$ , we move the  $1$  over  $T$  outside here.  
 That's not an issue.  
 And here, we write capital T very big,  
 because I explained we want to find a Fourier expression  
 for non-periodic function.  
 How we get a non-periodic function,  
 we write the period of the original periodic function  
 very large.  
 So we just make infinity here, because  $T$  is so big.  
 Likewise, you have minus infinity.  
 So here,  $e$  to the power minus  $i 2 \pi$ , minus  $i 2 \pi$ .  
 So  $n$  divided by  $T$ . And I argued  $n$  divided by  $T$   
 is nothing but  $u$ .  
 So the summation becomes integral.  
 And the discrete point becomes a continuous variable.  
 So when  $n$  is very big, you have many, many sampling points.  
 So this  $n$  divided by  $T$  becomes  $u$  here.  
 We've got  $T$ . So  $I$  of  $T dt$ .  
 So you've got this part.  
 This is a continuous  $x$  present.  
 So it looks very nice.  
 And the other side, it's just a copy down here.  
 So here, we utilize the relationship  
 $u$  equal to  $n$  divided by  $T$ . So here,  $u$  divided by  $T$ .  
 I mean,  $n$  divided by  $T$ .  
 So just call  $n$  divided by  $T$   $u$ .  
 So you can just get  $u$  here.  
 So any  $u$ , then you can recover.  
 So because when  $T$  is very big, all those sampling points,  
 it really becomes closer and closer, OK?  
 This way.  
 And then we can go step further.  
 So we call this whole transformation, this  $x$   
 present, as a function  $f$  hat  $u$ .  
 Because the integral is raised back to  $T$ .  
 So  $u$  is a parameter.  
 And the parameter has a frequency interpretation.  
 So this is a Fourier transform of  $f$  of  $T$ .  
 And you've got a function  $f$  hat of  $u$ .  
 So this is the coefficient of Fourier series.  
 Stay here.  
 Then you still do summation.  
 So you add all the Fourier components together.  
 Again, here,  $n$  divided by  $T$ , you call it  $u$ .  
 And the summation is converted into integral.  
 Because the summation happens here.  
 Many, many terms.  
 And each term is weighted by  $1$  over  $T$ .

And each functional value is value,  
 say, this arbitrary sampling point.  
 The functional value is here.  
 And this value is weighted by this small interval.  
 As  $T$  increases, and the  $\Delta u$  is  
 the interval between two sampling point becomes so small.  
 Therefore, this summation, really,  
 just each term is a functional value weighted by the interval.  
 And all these things added together  
 is nothing but an integral.  
 So the summation becomes an integral.  
 And the integrand is a Fourier transform  
 of the original function  $f$  of  $t$ , denoted as  $\hat{f}$  of  $u$ .  
 And the kernel here is  $e^{-i 2 \pi u t}$ .  
 Because  $n$  divided by  $T$  is  $u$ .  
 So you've got this one.  
 So you see,  $f$  of  $t$  can be expressed  
 as an integral of the spectrum of the original function.  
 So the kernel here without a minus sign.  
 But when you compute  $\hat{f}$  of  $u$ , you have a minus sign.  
 Otherwise, they look the same.  
 So this is so-called forward and inverse Fourier transform.  
 So given a function  $f$  of  $t$ , you do integral like this.  
 You've got a continuous Fourier spectrum here.  
 So this is a forward Fourier transform.  
 So given your Fourier spectrum, continuous Fourier transform,  
 and you can do integral or transform again  
 to recover the original function  $f$  of  $t$ .  
 So this is an inverse Fourier transform.  
 Now we can deal with a non-periodic function  $f$  of  $t$   
 in terms of Fourier spectrum.  
 We can go forward and backward.  
 So let me give you some picture to make the abstract formulation  
 a little more visible.  
 Suppose you have a function, rectangular gate function.  
 This  $\Pi$  of  $t$  is just like a gate, right?  
 So you have this gate function.  
 The amplitude is 1.  
 The total length of the interval is 1.  
 So area under the curve is 1.  
 Looks like a gate, and it's rectangular.  
 You can call it either way.  
 $\Pi$  of  $t$ , we will use this function quite often later on.  
 So this is a gate function.  
 This is a non-periodic function.  
 Based on what we learned in the previous lecture,  
 you do not know how to express this non-periodic function  
 as a summation of a sinusoidal component.  
 It doesn't matter.  
 You know how to do Fourier expression  
 for a periodic function.  
 So let's just make a periodic function  
 from the single gate function.  
 So we make a periodic period 8.  
 So you have multiple copies, infinitely many copies  
 of gate function.  
 Looks like what you have here.  
 This is a periodic function, and we  
 know how we can represent this continuous periodic function  
 as a summation of a sinusoidal component.  
 That's what Fourier series is used for.  
 You can just plug in the formulas I explained later on.  
 You've got a Fourier series expression.  
 The coefficient can be computed according to those formulas.



So you've got a coefficient for DC components  
and for the first harmonic component  
and the second component.  
And in complex form, they are symmetric for real function.  
So the Fourier series will be obtained  
according to coefficients we computed by those formulas.  
So this is the coefficient we will have, something like this.  
And the trick I mentioned, to find non-periodic function,  
to find the Fourier spectrum sinusoidal expression  
for a non-periodic function.  
And what we will do is just keep increasing  
period of the period function.  
So we make 8 becomes, say, 16, 32, 64,  
larger and larger.  
So for each given period  $t$ , no matter how large,  
you can compute it according to Fourier series theory,  
those formulas.  
And what's the difference?  
When period becomes larger and larger,  
the interval between adjacent Fourier components  
becomes less and less.  
So you see as  $t$  becomes larger and larger,  
so you got a denser and denser sampling picture  
in the frequency domain.  
What's the horizontal axis?  
So the horizontal axis is  $u$ .  
Remember I told you the horizontal axis is  $u$ ,  
defined as  $n$  divided by  $t$ .  
 $t$  becomes larger and larger, this interval  
gets smaller and smaller.  
So when capital  $T$  approaches infinity,  
so this Fourier spectrum really approach a limit  
that is a continuous function, even a loop of this function.  
So that's a geometrical picture of how  
you derive continuous Fourier spectrum from discrete Fourier  
spectrum.  
Discrete Fourier spectrum corresponds  
to periodic functions.  
And the continuous spectrum really  
corresponds to non-periodic functions.  
So the trick is to make capital  $T$  approaches infinity.  
So this is just a geometrical picture.  
So this is a one-slice summary.  
So given a one-dimensional function  $f$  of  $t$ ,  
you can perform this Fourier analysis.  
You got a Fourier spectrum.  
This is a continuous spectrum, but you  
can understand this spectrum as a limiting  
case of periodic function.  
So the geometrical picture ought to be kept in mind.  
So you have a very clear picture in the case of Fourier series.  
Now in the limiting case, you have  
continuous Fourier spectrum for a continuous one-dimensional  
function.  
So this is a so-called forward transform.  
And it's a part of this inward transform.  
So you have Fourier spectrum, spectral representation  
of the original function.  
Then the original function can be recovered  
by adding all these individual infinity many.  
But each component is so tiny.  
But you add all of them together.  
That's not summation anymore.  
And this is integral.

But integral, summation, basically the same idea.  
You add all these Fourier components all together,  
and then you recover the original function.  
This is the way we desired.  
So we want to represent the original function  
as a sum of many waves.  
Now, so many waves.  
Each wave component is tiny.  
But infinity many added together  
will give us a definite result. That's your original function.  
So for all these mathematical operations, meaningful.  
And I underlined in the previous class.  
So the periodic function ought to be square integrable  
over the interval 0 to 1, or over interval 0 to capital T,  
or minus capital T divided by 2 to positive capital T divided  
by 2.  
So all these operations should be convergent.  
So the mathematical definitions and derivations  
will be meaningful.  
In this case, we are talking about non-periodic function.  
So the similar assumption applies here.  
That means  $f$  of  $t$  is non-periodic.  
But if you do squared integral over the whole number of sides,  
so the function of  $f$  of  $t$ , although non-periodic,  
but the squared function should be integrable.  
That means the integration of the squared function  
should be some finite number.  
If you think about a function like a constant function,  
and you just do integral squared or not,  
you've got an infinite number.  
Then you have a divergent problem.  
So mathematical operation no longer meaningful.  
So this is a summary.  
And this summary, and I would say,  
try to precede the beauty behind this formulation.  
You see, this is a function.  
And this is a Fourier spectrum.  
By inner product, this function really  
projected to infinity many basis functions.  
So you've got all the coefficients.  
Just like you represent a 3D function in terms  
of  $x$ ,  $y$ ,  $z$  components.  
That's the same thing.  
So you have this Fourier expression.  
The unit vectors, really the basis functions,  
sinusoidal functions, in this form.  
So geometrically, the same idea.  
Then once you have this expression,  
and then you can recover the original function  
by adding all these small  $x$ ,  $y$ ,  $z$  components together,  
you get an original three-dimensional vector.  
In this case, you add all these small sinusoidal functions.  
This is coefficient for a particular sinusoidal function,  
 $e$  to the power  $i 2 \pi u t$ .  
You add all these tiny components together.  
You recover the original function,  $f$  of  $t$ .  
So we say you can go back and forth  
without information loss.  
Then the  $f$  of  $t$  is equivalent to height  $f$  of  $u$ .  
That's Fourier transform of original function.  
The equivalency really means almost everywhere.  
If you have piecewise continuous function,  
at this continuous point, say you  
have finitely many discontinuous points,



And if you just look at appendix of some imaging textbook, so you see those formulas and you plug in and you can compute. That's not the most important. What I have been trying my best to explain, I really want you to visualize what's going on so you know the ideas behind all these computations.

So the essential part already covered summarized here. We can represent a non-periodic function in terms of its continuous Fourier spectrum. And with the continuous Fourier spectrum we can recover the original function. So this is just a complex wave and weighted by this coefficient. And certainly this coefficient is small. I'm not saying the amplitude of this  $f$  is small. Really this amplitude is weighted by  $\delta u, du$ . So this whole thing is a small weighting factor. And this integral is really just a sigma sign. Add all these components together and remember the cartoon and a few students hold small blackboard. The waves added together recover the original picture. So that is the illustration of the idea behind Fourier transform forward and inverse. So this inverse Fourier transform is just the summation process to recover the original function. So understand the idea. Then you do the computation. Otherwise you just do computation. You feel you're like a robotic, like a computer. We are human beings. We want to visualize things so that you can imagine new applications and new results.

The second part, we give you some examples. So if you understand the formulas I showed you on the previous slides, the rest are just somehow straightforward. So gate function. The gate function, we already defined the gate function. It's just a standard gate. And the area and the curve is one. So what is the continuous Fourier spectrum of the gate function? We kind of already showed you, but now make the process formal. So the continuous Fourier spectrum is nothing but the integral that I derived for you in the first part. So you do the integral, you have limits, infinity, negative infinity, because we write capital  $T$ , really big. So that's just infinity. Then you have inner product, original function projected onto this complex basis function, indexed by continuous variable  $u$  here. Can you see here?  $2\pi u t$ . For giving you, you do this inner product once. Inner product is nothing, just projection. You just do computation. The example was chosen to make our life easier. So this is the function of value 1 over this finite interval. So Fourier transformation becomes this expression. And you just solve this equation a few steps, and you have this sine  $\pi u$  divided by  $\pi u$ . So this is for unit interval. And for general interval, so with period capital  $T$ , and you got a more general sinusoidal

I mean this whole function

is called sinc function. You got a more general sinc function. You can just try to find Fourier transform for this more generalized gate function. But for standard unit gate function, you have these results, but you can verify. And this function,  $\text{sinc } \pi u$  divided by  $\pi u$  is very important, not only for gate function itself. And we will utilize the Fourier transform of this function in the next lecture. So let's just look at what is the functional form of this result. This result has a name called sinc function. It's not sine function. Sine function is just the sinusoidal vibration. But this sinusoidal vibration is divided by  $\pi u$ . You're getting larger and larger, so you can imagine the amplitude getting smaller and smaller. But when  $u$  is proportional to  $\theta$ , and  $\text{sinc } \pi u$  proportional to  $\theta$ ,  $\pi u$  proportional to  $\theta$ , then you use your calculus 1. You find the limit using something called the L'Hopital rule. So you can find the limit is 1. So this is the functional form of sinc function, something like this. So this is the first example. What is the continuous Fourier spectrum of a standard gate function? And if you like, you can find a non-standard gate function. The period is not 1. And what will be the sinc function for that gate function of period capital T? This is the first example. Second example, so suppose the non-periodic continuous function is a triangular function or triangle function. This is a triangle function defined this way. What is Fourier transform? So we want to represent this triangle function with many, many one-dimensional waves. Add them together, you can recover this triangle function. And computation is not so much different from what we explained. We plug in the triangle function into the formula for Fourier transformation. So you have this inner product. So the integral from minus infinity to infinity. This is a function. Then you have this basis complex, basis function,  $e^{-i 2 \pi u x}$ .

In my book, I call it  $U$ . But here, this example I copied from the Stanford textbook. They call it  $S$ . It's not a function of  $T$ . It's a function of  $X$ . It's really variable. It doesn't matter. So in this case, you have the frequency as  $S$ . So  $S$ , the Fourier transform is performed with respect to  $X$ . And just keep doing the trivial computation step by step. And what you have is sinc squared. Then this is  $U$ . Here is  $S$ . So sinc squared. So that's just somehow this. See, this is the gate function. The Fourier transform is a sinc function. So here is a sinc function. And for this gate function, you got sinc squared. So you have many examples. So any function you just plug into the Fourier transformation. And you just do the analytic computation.

And sometimes you are lucky. You can find closed form solution. And many times, you cannot find closed form analytic solution. In that case, you can do numerical computation. You can still find Fourier transform in terms of numerical values. And a few more examples shown here. They are nice. So you have this Gaussian function.  $E$  to the power  $\pi t^2$ . So you have this bio-shaped Gaussian function. And this function is certainly square integrable. Then you can perform Fourier transform upon this Gaussian function. The Gaussian function is elegant. The Fourier transform, you plug it into Fourier transform formula. You do the analytic computation. It's Fourier transform. It's still a Gaussian function. So in the time domain, the variable is  $t$ . In the frequency domain, the variable is  $u$ . And I use  $u$  to index the frequency component. And if you use  $\sigma$ , or use  $s$ , or use what? You check the textbook website and the different versions of Fourier transforms are available. But you really need to see what's going on. That's in the product that you compute. And you have one variable in one domain. Time or space or whatsoever. And in the reciprocal space, you have frequency  $\omega$ ,  $f$ ,  $u$ ,  $s$  whatsoever. But this is just a way to represent the original signal in terms of sinusoidal wave functions, different kinds of wave components added together. And now let's look at a very ridiculous case. It's  $c$ . So  $c$  is constant. And I have been saying to write Fourier transformation, all the mathematical computations, derivations, make sense. We ask function, remember the  $L^2$  space, that means function, you do the integral, you perform the inner product, we make sure the Fourier coefficient converges to some finite number. And the  $c$  constant, if  $c$  is not  $0$ , and this constant function is not square integrable. So you do Fourier transformation upon this kind of function, you have to be very careful how you interpret this some generalized Fourier transform. So in the textbook I explained, we try to understand how you perform Fourier transform for constant non-periodic function. And for some generalized function,  $\delta u$  or  $\delta x$ ,  $\delta t$ , those things are not easy to explain. But indeed I gave some explanation in the book chapters. And so if you are interested, you can read. But here you can just understand as just some rigorous mathematical derivation in the sense of distribution. And you can have this relationship constant, the Fourier transform is a delta function. The delta function, the Fourier transform is a constant. So read the chapter for detail. But mathematical regular is not stressed here. And for delta function, you perform a translation. So by amount of  $a$ , so if  $a$  is equal to  $0$ , so the delta function Fourier spectrum is a constant. So all the frequency components need to be added

together to make this delta function. But if you have a shifted delta function, the Fourier spectrum is shifted. So the phase shifting factor depends on frequency. So for higher frequency, you have a larger phase shift. For lower frequency, you have a smaller phase shift. And we will talk about this phenomena in the next part about properties of Fourier transform. So right now you're just thinking these are a bunch of examples. Some are easier. Like a gate function, triangle function, Gaussian function, these functions are nice. So you do integral, the functions are square integrable. So you can understand the existence and so on using your conventional mathematical sense. But when you have some generalized function, some function not square integrable, like constant delta function, you need to use deeper mathematics. And I explained a little bit in the chapter, but I'm not going to explain it here. So much for examples. If you like, you can find more examples. Now we discuss properties of Fourier transform. The Fourier transform as I argued has been well motivated, clearly defined. And what are the properties of Fourier transform? They have a bunch of very nice properties. So first, Fourier transform, you give an arbitrary function  $f$  of  $t$  as an input through a box, black box, called, it's not a black box, you should know everything. It's a white box. Through this box or system called Fourier transform, what's the output? Output is Fourier spectrum of the input function. Then we can ask if this system or the transform is linear or not. And we learned the linear system theory, so any system we can always ask this question. And the answer is positive. So if you have function  $f$  and function  $g$ , you perform Fourier transform. You got Fourier transformation for  $f$ , Fourier transformation for  $g$ , again, and a lot of versions of Fourier transform. Here just copied from Wikipedia. If you really prefer unified treatment, read my chapter. Here just show you different things. So do not stick with one notation, one set of notations. So here, so the Fourier transformation is really linear. So that means if you add input function  $f$  and the input function  $g$  together, or you linearly combine them together, the combination, the Fourier transformation of the combined  $f$  and  $g$  linearly combined version, in the Fourier domain, you see the same linear combination of their corresponding Fourier spectrum. So translation property, and I mentioned that this is the previous slide for delta function. For arbitrary function, so you have  $f$  is the original function. Then you have its Fourier transformation, height  $f$ , Fourier component  $\sigma$ . If you do translation in one domain, then you have phase vector weighted upon the Fourier transform. You have modulation. Modulation seems just like the other way. You translate it in Fourier domain. You go back with this weighting factor in the spatial

or temporal domain. And the scaling, so you just say you have original function  $f$  of  $t$  or  $f$  of  $x$ . You have Fourier transform. Then what if you just scale  $f$  of  $t$  or  $f$  of  $x$  with a factor  $a$ ? Then is  $f$  of  $ax$ , and what will be Fourier transform? And the conjugate and so on. So all these are properties you can read. Let's look into more detail. Okay, the first linearity. So this is just, again, this part copied from Stanford textbook. So you have a function  $f$  and  $g$ . There are Fourier transformations. So you have Fourier transformation. You have this summation, the summation of the original function. Then you perform Fourier transformation. The result is the same as summation of Fourier transformation of  $f$  and Fourier transformation of  $g$ . So this is additivity. So in the system, linear system lecture, we explain that. How about the scaling or homogeneity? So if  $f$  is scaled by  $\alpha$ , then we perform Fourier transform. That's the same as you perform Fourier transform of  $f$ . Then you scale the result by the same scaling factor. So you can verify these two properties according to the formula by defining the Fourier transform. So you can just see here, just as an example, you show additivity. And you can similarly show the scaling property. And for safety property, again, and all these properties, you have the definition. You plug in the translation or scaling or whatever. You plug any modification into the definition. Then you show what is your outcome in terms of the Fourier transform of original function. Then you find the relationship. You call it the corresponding property. In this case, we are talking about safety property. So you save the variable  $t$ , save the  $f$  of  $t$ . So the variable  $t$  by  $b$ . So you have this. You perform this substituting transform,  $u$  equal to  $t$  minus  $b$ . So make sure this  $f$  becomes  $f$  of  $u$ . Then you just change the limit. You just do step by step derivation. Then you got this original Fourier transform. And you have this phase factor, factorized out. So you know if you have original  $f$  of  $t$  corresponding to Fourier transform capital  $F$ , then if you do translation by  $b$ , then you have, by  $b$  in the time domain, then you have phase factor in the frequency domain. So these derivations are not complicated. So you can review just a way to remember the property and also a way to familiarize yourself with Fourier transform. Scaling, I mentioned scaling. So scaling property really needs to be discussed in two cases. One case is a greater than zero. The other is a less than zero. If  $a$  equal to zero, that doesn't make sense because originally you have a function  $f$  of  $t$ . So  $t$  will change from minus infinity to positive infinity. If you change



$t$  to zero, that means you turn one dimensional function into a single point. That doesn't make sense. So we don't want  $a$  equal to zero. So when  $a$  is greater than zero, you do the derivation. You see the original function,  $f$ , then you have the Fourier transform capital  $F$ , or you just call the height  $f$ , whatever notation you choose. So if you make the function scaled in terms of variable  $t$ , so  $t$  becomes  $a t$ , then the Fourier transform will become  $1$  over  $a$  times original function, but the original frequency component becomes  $s$  over  $a$ . So see the variable in time domain,  $a t$ , the Fourier domain becomes  $s$  over  $a$ . So the factor in one space is  $a$ , in the other space is a reciprocal of the scaling factor. So this is for  $a$  greater than zero. And for  $a$  minus than zero, the mathematical reason is between the lines. So the resultant Fourier transform is minus  $1$  over  $a$  here, then still same thing. So the difference is minus sign. Because you already said  $a$  is negative, so minus  $1$  over  $a$  is the same thing as  $1$  over absolute value  $a$ . So summarizing two cases together, so we say  $f t$  has a Fourier transform capital  $f$  of  $s$ . And in my book I call it height  $f u$ . So you have a pair of Fourier transform here. Then if you scale variable  $p$  by factor  $a$ , then the Fourier transform of original function will be modified with scaling factor  $1$  over absolute value  $a$  and original Fourier spectrum. But the variable is weighted by  $1$  over  $a$ . And inside this parenthesis, you do not have absolute sign. So this is just scaling property. And again, I like visualization. So visually, what does scaling property mean? So look at this picture. This is the top part is original function. Then bottom part is Fourier transform, color coded. So for this blue triangle function, you have this light blue Fourier spectrum. So the blue triangle is kind of a sub. In the Fourier domain, it's kind of spread out. So when you use scaling factor to scale this function, so you turn this blue function into the same shape, but just the horizontal axis or coordinate is deleted a little bit. So you have a little fighter triangular function, this orange function. Then the Fourier spectrum, because in the  $T$  domain, the function got a little fatter. And the frequency spectrum got a little bit slimmer. So you get narrower. So you make the triangular function even wide. So you got quite a sub peak in the frequency domain. So the two domains, they are really closely related. They have duality. So you have very narrow in one space, you will have very broad profile in the other space. So if you have a delta function,

it's very narrow in one space. You have constant, very flat, wide in the other space. And the sweet spot is Gaussian function. So you have Gaussian function in one domain, and you have Gaussian function in the other domain. It can be the same. So this is scaling property and visualization of the property. And you have many, many properties. So this is derivation. So you can ask many questions, and you consider different operations you could perform on the original function. Say you have original function  $f$  of  $t$ . The Fourier transform is height  $f$  of  $u$ . If you do derivative operation upon original function  $f$  of  $t$ , what will be the Fourier transform of the derivative of the original function? So things like this, you can keep asking many, many interesting questions. And you can just plug in the fundamental operation, in this case, derivation, into the Fourier transform formula. Then you do derivation step by step, and you will have the relationship. Say the Fourier transform of derivation of function  $f$ . What's a Fourier transform? Original Fourier transform before derivation is capital  $F$  of  $s$ . So with derivation, you have this factor,  $2\pi i s$ . So you have this phase vector, as a weighting vector, upon original Fourier transform. So you do derivation in time domain. That is multiplication in frequency domain. The multiplicative factor is frequency dependent. For different frequency, you have different value, and it's linearly proportional to the frequency component. So that's derivation. Very important Fourier transformation pair is called train of impulse function. Because the train of impulse function looks like a comb, also paired combs, we call. So you have many delta functions with period  $\Delta t$ . So the  $n$  equal to  $0$ , you got delta function at the system origin.  $n$  equal to  $1$ , so you got  $t$  minus  $\Delta t$ , so you have delta function. Right safety by period  $\Delta t$ . So you have many of these. So you have a train of delta functions in time domain. We will utilize this slice extensively in the next lecture. The question here is, if you perform Fourier transformation of this comb function, a bunch of delta functions, a series of impulses in the  $t$  domain, what is the Fourier transformation, Fourier transform of this time domain comb? And again, it's just a comb, but with different period. The period in  $t$  domain is  $\Delta t$ . In the frequency domain is  $1$  over  $\Delta t$ . This is consistent to the scaling property. Imagine that I make one thing, a factor in one domain, then the corresponding original function of spectrum will be a factor. So narrow in one domain, factor in the other domain. Here the period is smaller in one domain. The period becomes larger in the other domain. Really just similar thing. But here we are involving

delta functions, generalized functions. So the relationship is shown here as a tool you can utilize. But why we have this relationship? And when we deal with delta functions, deal with functions that is not square integrable, will the result still mathematically meaningful? And that is what I mentioned involves deeper mathematics. And they are rigorous, but I don't have time to explain in detail. If you are interested again, read the chapters. I try to make BME version of the explanation in what sense these formulas holds will not give you absurd results.

Next theorem is probably most important property of the Fourier transform. And also very important property for linear system theory is called the convolution theorem. And the convolution we already learned is something, remember example, you have one vector or one function. You have a second function. You flip one, then you match together, just add a parcel product together. And you need to keep doing safety, get a bunch of numbers and these numbers put in a list. That's the result of you discrete convolution. And in continuous case, you have two continuous functional forms. You flip one, then you translate the functions continuously over interval. And for each translation, you match two functions together, that's multiplication. The multiplication is integrated with respect to the variable. And this whole thing, whole process is called convolution. Why convolution is important? Because linear system theory is important. Once you know safety invariant linear system has an impulse function, either continuous impulse response function, either continuous or discrete. Once you know that for arbitrary input through convolution, you can find output of the system. So from input to output and that's a forward process that we do routinely in engineering practice and in other fields, many other important research fields as well.

So this is convolution. I also mentioned that convolution is advanced form of multiplication. Now we can make the statement very specific. Actually convolution in  $T$  domain is multiplication in Fourier domain. So they are equivalent. So you can perform convolution as I told you in your homework, in the class. You can continue to do convolution. Or you don't do it. You just for the given functions,  $f$  and  $g$ , you find Fourier transform for  $f$ , Fourier transform for  $g$ , you multiply them together. Just the standard multiplication in the Fourier domain. This product is Fourier transform of convolution. So to find convolution, you can do multiplication then you perform inverse Fourier transform. You get exactly the same result. So this is just the geometric interpretation of convolution theorem. So let me give you a visual example. Remember the gate function. You do Fourier transform. What do you have? Remember you got a single function. Then I give you a second example, a triangle function. You can think two gate functions. You do convolution, two gate functions.

Convolution of two gate functions gives you a triangular function. I just told you the gate function is a single function in Fourier domain. So the convolution in spatial domain, two gate, you got a triangle. So in Fourier domain, the two gate functions are two single functions. So you multiply them together. A single function squared, that is Fourier transformation of the triangle function. That is what you show.

So here you know why. Because this is convolution theorem. So why do we have convolution theorem? Just plug in convolution operation in terms of the definition.

The two functions, F and G, you do multiplication together. You just plug in the definition of Fourier transform. You keep doing things, make it in the form of convolution. So you can go from multiplication in one domain to the convolution.

To this form, the convolution in time domain or space domain, then you perform Fourier transformation of the convolution result. That is nothing but multiplication of Fourier transform. So just the derivations, step by step, and then you can review after class, put a cup of coffee or tea, and just go through. So this is derivation of why you can prove. Let me give you some deeper insight. You may or may not understand, but this is the way researchers are encouraged to do. You try to understand those trivial mathematics with some visualized picture. So why you have this? So you can actually think verbally. It's just my understanding. You cannot find from the internet. Let me make some comments. So convolution is a result

we derived, a process we derived for safe environment linear system. So I believe you can show for safe environment linear system. So you gave the system a sinusoidal input. Then what will be output? The output will be also sinusoidal. And also for safe environment linear system, you gave sinusoidal input to the system. The output must be sinusoidal or sinusoidal function at the same frequency.

So linear environment system would not give you anything, say, higher frequency, lower frequency, and it will not change the form of input function if the input is sinusoidal. This can be shown.

So in other words, it says convolution in T domain must be multiplication in Fourier domain. So I jump a few steps. So think about this. You have a sinusoidal input to linear system. The output is still sinusoidal, but it may change some factor scaled upon the original thing.

So there must be some specific number at that frequency multiplied to give you output for this sinusoidal function. So for arbitrary function, you decompose into many, many sinusoidal function. Add it together. Then put it through the system. For each particular frequency, you just weight the frequency with a frequency-specific number. So that is multiplication in Fourier domain.

And that particular frequency-dependent number must be Fourier transform of your impulse function. So the HT, you do Fourier

transformation. You got a Fourier spectrum with all the coefficients or weighting vectors, frequency-specific. So why convolution is multiplication in Fourier domain? Just this sentence. So you understand what's going on. And so I made a claim. So you have sinusoidal function at a certain frequency. You input into linear system, safety environment linear system. The output is invariability. So the certain functional form put into the system, the output is still same form of the function. And the only sinusoidal has this property. You have delta function is delta. Output generally is not delta. It's something like this in your homework. So only sinusoidal has this property. So if you show, I think you can show this. If you show this, you can make further statement. So you think about convolution in one domain. And you can ask this question. Convolution in one domain. And in what domain? You can think about Fourier transform, wavelet transform, Hadamard transform, many, many transform. Each transform is just different way. Orthogonal presentation is not unique. Because also normal basis is unique. You have infinitely many transforms. And any other transform will also have this convolution theorem. The answer is no. Because only sinusoidal function has this safe shape invariability. Does not change shape. Does not change frequency. And the second paragraph really leads to an important conclusion. And the convolution theorem only holds for Fourier transform. If you have some X transform, it's not a Fourier transform. You do not have convolution theorem. So this is something you do Google search. You say characterization of convolution theorem. There are multiple mathematical papers talking about this characterization. Indeed, the convolution theorem only holds for Fourier transform. Why? You read pages, pages, mathematical derivation. But really just visualize it. Just these few comments I have. If any of you interested in mathematics, any of you, I do not expect all of you understand. Any of you really understand my comments. You want to write a technical report, a small paper, just so that you can discuss with me. This is a heuristic view. View of convolution theorem. You have next important property. Passive identity. This is, again, I copied from Stanford textbook. I have a very good textbook. I used this student in the previous semester. Now I just have my book draft so I didn't make things too complicated. Because that textbook is quite safe. It's for whole-sized Fourier analysis. This is the identity. In T domain, squared function, you just integrate them together. What is this? I keep saying to make classic mathematical sense, function  $f$  of  $t$  must be square integrable. If not integrable, you do not have a finite number. You got divergent behavior, not easy to deal with. Not to say impossible, but much harder. You need to use deeper mathematics, like distribution and so on in some mathematical sense. But if you have this converging behavior, much easier to understand.

Alternating current, the light,  
what's the power? Power is just that you have oscillating  
curve in the sinusoidal waveform  
or any waveform. So it's just current squared.  
That is, suppose you have unit resistance.  
This is the measure of energy.  
You do this integral, left-hand side is just total energy.  
This is total energy in time domain.  
The right-hand side, you have frequency components.  
Suppose the waveform, the current, is arbitrary.  
You can treat that as summation of many sinusoidal  
OC circuit or OC current, but at a different frequency.  
But for each given frequency, you have AC  
alternating current. What's the power? For a given frequency,  
the power is proportional to the amplitude squared.  
This is the total power you see  
in the time domain. So this is the straightforward high school  
circuit teaching. Tell you for current  
at a given time, squared times R, that's power.  
For different time, you add all these together. That's total energy  
the resistor consumes. In terms of Fourier analysis,  
you think that arbitrary waveform is really  
a number of many, many sinusoidal components.  
For each component, the energy is amplitude squared  
and all these sinusoidal AC current  
added together, the frequency expression.  
And this property basically says energy is  
conserved. That's clear. That's the  
physical perspective. What's the geometrical perspective?  
So I told you earlier, I tried to encourage you  
to think abstractly. So a vector, n-dimensional  
vector, is a point in an n-dimensional space.  
Then we say the function  $f$  of  $t$  is a vector, just an infinity  
dimensional vector. And you use a vector to  
approximate a continuous function,  $f$  of  $t$ . So  $f$  of  $t$   
is a vector, infinity dimensional vector,  
is a point in infinity  
dimensional space. And what is the length? What is the length  
of that vector or that function?  
The length is computed by squares, individual  
components. Individual components squared, then you  
add them together. You do square root. So you put square root  
so the left-hand side is the total length of  
vector or function  $f$  of  $t$ . So that's  
the total length. And the Fourier transform, I say, is also normal  
transformation. So you just think this  $f$  of  $t$   
in our original coordinate system is  $f$  of  $t$ .  
And you put it in Fourier space.  
Fourier space is really also normal transformation.  
It's a coordinate system rotation. But this is just not  
as easy as  $xy$  rotation. You rotate it to  $t$  and  $s$ .  
But the geometrical idea is the same. It's just the  
infinity dimensional space. You do a rotation called Fourier  
transformation. Then in the Fourier domain, this  
function becomes  $\hat{f}$  of  $s$ .  
So each component you squared, then added together.  
What's the geometrical meaning of right-hand side? If you put square root  
here, that's the length of the vector in the  
Fourier space. So geometrical meaning is that  
the total length doesn't change. So this is  
what I explained here. So you have geometrical understanding  
why you have this identity. So certainly  
you can just go through the formula. And the formulas  
make things rigorous. And we like doing derivation.

But I always think for complicated things, if you understand, just you can visualize things. And that is better, higher capability than you can just follow the derivation. Oftentimes you follow derivation, but you really don't know what's going on. So the geometrical understanding, you can visualize the process. And that is the place, that is the time you are getting better understanding. So far we have been talking about one-dimensional transformation. And you can extend Fourier transformation from one-dimensional to two-dimensional. So the wave component is no longer one-dimensional wave, rather just two-dimensional wave along different directions. You can go along  $x$ , along  $y$ , diagonally, whatsoever. So this is a picture, very noisy picture. You perform Fourier transformation, got this two-dimensional Fourier spectrum. We know high frequency components. Those wave oscillate at high frequency. Oftentimes not very useful. Those components are noise. So in Fourier domain, we just zero out all these things. You see, perform inverse Fourier transformation, you get noise removed. So this is why say in Fourier domain, sometimes it's easier. And this low pass, high pass filter, original image, you can use low frequency components like this. And you can just zero out low frequency components, keep a certain range of high frequency components. That gives you edges of the building. So that's another example. And for rectangular, 2D rectangular, 2D impulse function, you can do analytic computation. And in high-dimensional space, you have something unique called rotation property. Rotation property along one-dimensional direction, you cannot rotate. You can flip. You cannot rotate. But two, three-dimensional, you start having rotation property. So this is a new thing, dimensionality dependent. What's rotation property? Basically it says you have 2D function here. You have 2D Fourier spectrum here. If you rotate the 2D function this way, the Fourier spectrum will be rotated the same way, by same amount of angular change. And why you have this thing in any dimensional space? So this is mathematics. You can follow through. Not very heuristic. Let me give you a heuristic explanation. So you have a function. You have Fourier spectrum. So Fourier spectrum, each point is a wave component moving in certain direction, just the wave. So all these wave components added together give you the original function. Now you rotate the original function by a certain angle. And all the original wave components are now going this way, the function like this. The original function rotated by that direction. So these wave components, to approximate that function, should be rotated by same angle. So this is geometrical picture. Why you have this rotation property of Fourier transform? And you can just follow through the mathematics to be sure. But this is geometrical picture. So I hope you start understanding the idea I'm trying to convey. The foundational part is very important. And Fourier analysis and linear system are not only important for imaging, for engineering. Physics and mathematics

