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An Interactive Guide To The Fourier Transform

by Kalid Azad · 287 comments

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The Fourier Transform is one of deepest insights ever made. Unfortunately, the meaning is buried within dense equations:

$$X_k = \sum_{n=0}^{N-1} x_n \cdot e^{-i2\pi kn/N}$$

$$x_n = \frac{1}{N} \sum_{k=0}^{N-1} X_k \cdot e^{i2\pi kn/N}$$

Yikes. Rather than jumping into the symbols, let's experience the key idea firsthand. Here's a plain-English metaphor:

- What does the Fourier Transform do? Given a smoothie, it finds the recipe.
- How? Run the smoothie through filters to extract each ingredient.
- Why? Recipes are easier to analyze, compare, and modify than the smoothie itself.
- How do we get the smoothie back? Blend the ingredients.

Here's the "math English" version of the above:

• The Fourier Transform takes a time-based pattern, measures every possible cycle, and returns the overall "cycle recipe" (the strength, offset, & rotation speed for every cycle that was found).

Time for the equations? No! Let's get our hands dirty and *experience* how any pattern can be built with cycles, with live simulations.

If all goes well, we'll have an aha! moment and intuitively realize why the Fourier Transform is possible. We'll save the detailed math analysis for the follow-up.

This isn't a force-march through the equations, it's the casual stroll I wish I had. Onward!

From Smoothie to Recipe

A math transformation is a change of perspective. We change our notion of quantity from "single items" (lines in the sand, tally system) to "groups of 10" (decimal) depending on what we're counting. Scoring a game? Tally it up. Multiplying? Decimals, please.

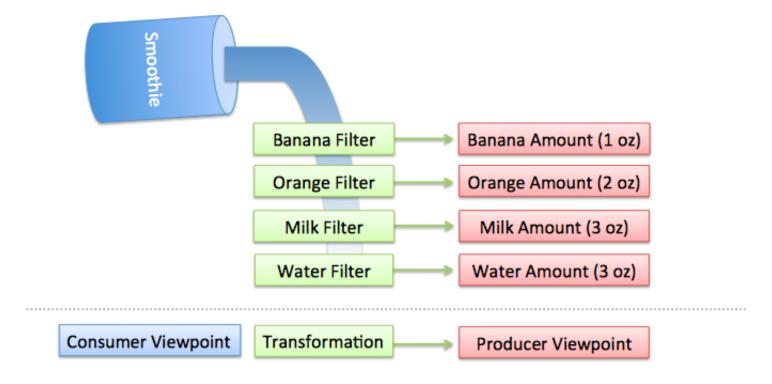
The Fourier Transform changes our perspective from consumer to producer, turning *What did I see?* into *How was it made?*

In other words: given a smoothie, let's find the recipe.

Why? Well, recipes are great descriptions of drinks. You wouldn't share a drop-by-drop analysis, you'd say "I had an orange/banana smoothie". A recipe is more easily categorized, compared, and modified than the object itself.

So... given a smoothie, how do we find the recipe?

Smoothie to Recipe



Well, imagine you had a few filters lying around:

- Pour through the "banana" filter. 1 oz of bananas are extracted.
- Pour through the "orange" filter. 2 oz of oranges.
- Pour through the "milk" filter. 3 oz of milk.
- Pour through the "water" filter. 3 oz of water.

We can reverse-engineer the recipe by filtering each ingredient. The catch?

- **Filters must be independent**. The banana filter needs to capture bananas, and nothing else. Adding more oranges should never affect the banana reading.
- **Filters must be complete**. We won't get the real recipe if we leave out a filter ("There were mangoes too!"). Our collection of filters must catch every last ingredient.
- **Ingredients must be combine-able**. Smoothies can be separated and re-combined without issue (A cookie? Not so much. Who wants crumbs?). The ingredients, when separated and combined in any order, must make the same result.

See The World As Cycles

The Fourier Transform takes a specific viewpoint: What if any signal could be filtered into a bunch of circular paths?

Whoa. This concept is mind-blowing, and poor Joseph Fourier had his idea rejected at first. (*Really Joe*, even a staircase pattern can be made from circles?)

And despite decades of debate

(http://wiki.answers.com/Q/What_is_the_history_of_fourier_series) in the math community, we expect students to internalize the idea without issue. Ugh. Let's walk through the intuition.

The Fourier Transform finds the recipe for a signal, like our smoothie process:

- Start with a time-based signal
- Apply filters to measure each possible "circular ingredient"
- Collect the full recipe, listing the amount of each "circular ingredient"

Stop. Here's where most tutorials excitedly throw engineering applications at your face. Don't get scared; think of the examples as "Wow, we're finally seeing the source code (DNA) behind previously confusing ideas".

- If earthquake vibrations can be separated into "ingredients" (vibrations of different speeds & strengths), buildings can be designed to avoid interacting with the strongest ones.
- If sound waves can be separated into ingredients (bass and treble frequencies), we can boost the parts we care about, and hide the ones we don't. The crackle of random noise can be removed. Maybe similar "sound recipes" can be compared (music recognition services compare recipes, not the raw audio clips).
- If computer data can be represented with oscillating patterns, perhaps the least-important ones can be ignored. This "lossy compression" can drastically shrink file sizes (and why JPEG and MP3 files are much smaller than raw .bmp or .wav files).

• If a radio wave is our signal, we can use filters to listen to a particular channel. In the smoothie world, imagine each person paid attention to a different ingredient: Adam looks for apples, Bob looks for bananas, and Charlie gets cauliflower (sorry bud).

The Fourier Transform is useful in engineering, sure, but it's a metaphor about finding the root causes behind an observed effect.

Think With Circles, Not Just Sinusoids

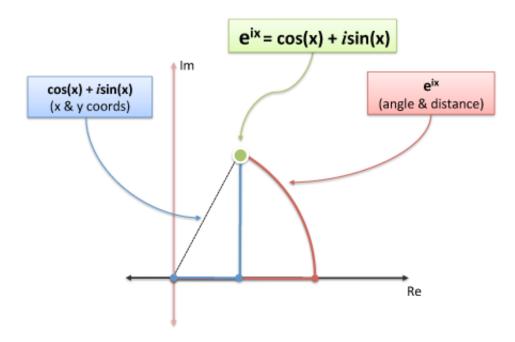
One of my giant confusions was separating the definitions of "sinusoid" and "circle".

- A "sinusoid" is a specific back-and-forth pattern (a <u>sine</u> (http://betterexplained.com/articles/intuitive-understanding-of-sine-waves/) or cosine wave), and 99% of the time, it refers to motion in one dimension.
- A "circle" is a round, 2d pattern you probably know. If you enjoy using 10-dollar words to describe 10-cent ideas, you might call a circular path a "complex sinusoid".

Labeling a circular path as a "complex sinusoid" is like describing a word as a "multi-letter". You zoomed into the wrong level of detail. Words are about concepts, not the letters they can be split into!

The Fourier Transform is about circular paths (not 1-d sinusoids) and <u>Euler's formula</u> (http://betterexplained.com/articles/intuitive-understanding-of-eulers-formula/) is a clever way to generate one:

Two Paths, Same Result



Must we use imaginary exponents to move in a circle? Nope. But it's convenient and compact. And sure, we can describe our path as coordinated motion in two dimensions (real and imaginary), but don't forget the big picture: we're just moving in a circle.

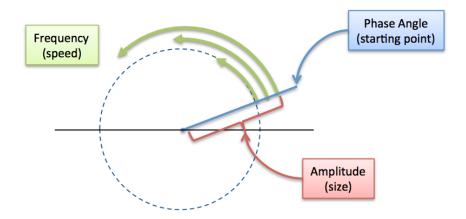
Following Circular Paths

Let's say we're chatting on the phone and, like usual, I want us to draw the same circle simultaneously. (*You promised!*) What should I say?

- How big is the circle? (Amplitude, i.e. size of radius)
- How fast do we draw it? (Frequency. 1 circle/second is a frequency of 1 Hertz (Hz) or 2*pi radians/sec)
- Where do we start? (Phase angle, where 0 degrees is the x-axis)

I could say "2-inch radius, start at 45 degrees, 1 circle per second, go!". After half a second, we should each be pointing to: starting point + amount traveled = 45 + 180 = 225 degrees (on a 2-inch circle).

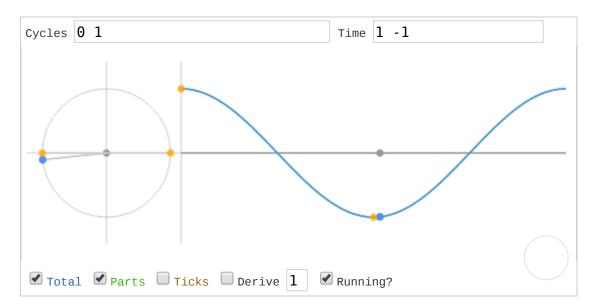
Describing A Circular Path



Every circular path needs a size, speed, and starting angle (amplitude/frequency/phase). We can even combine paths: imagine tiny motorcars, driving in circles at different speeds.

The combined position of *all the cycles* is our signal, just like the combined flavor of *all the ingredients* is our smoothie.

Here's a simulation of a basic circular path:



(Based on this animation (http://treeblurb.com/dev_math/sin_canv00.html), here's the source code (https://gist.github.com/kazad/8bb682da198db597558c). Modern browser required. Click the graph to pause/unpause.)

The magnitude of each cycle is listed in order, starting at 0Hz. Cycles [0 1] means

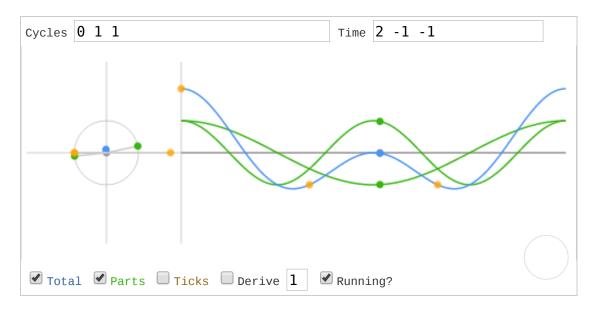
- 0 strength for the 0Hz cycle (0Hz = a constant cycle, stuck on the x-axis at zero degrees)
- 1 strength for the 1Hz cycle (completes 1 cycle per time interval)

Now the tricky part:

- The blue graph measures the *real part* of the cycle. Another lovely math confusion: the real axis of the circle, which is usually horizontal, has its magnitude shown on the vertical axis. You can mentally rotate the circle 90 degrees if you like.
- The time points are spaced at the fastest frequency. A 1Hz signal needs 2 time points for a start and stop (a single data point doesn't have a frequency). The time values [1 -1] shows the amplitude at these equally-spaced intervals.

With me? [0 1] is a pure 1Hz cycle.

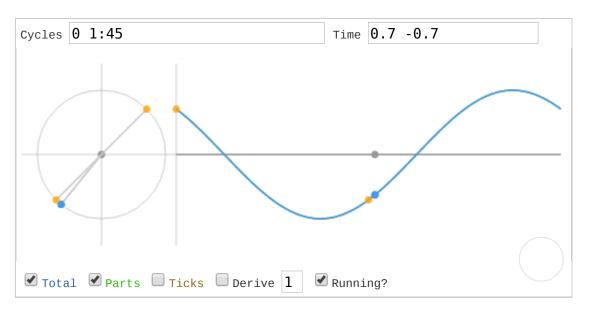
Now let's add a 2Hz cycle to the mix. [0 1 1] means "Nothing at 0Hz, 1Hz of strength 1, 2Hz of strength 1":



Whoa. The little motorcars are getting wild: the green lines are the 1Hz and 2Hz cycles, and the blue line is the combined result. Try toggling the green checkbox to see the final result clearly. The combined "flavor" is a sway that starts at the max and dips low for the rest of the interval.

The yellow dots are when we actually measure the signal. With 3 cycles defined (0Hz, 1Hz, 2Hz), each dot is 1/3 of the way through the signal. In this case, cycles [0 1 1] generate the time values [2 -1 -1], which starts at the max (2) and dips low (-1).

Oh! We can't forget phase, the starting angle! Use magnitude:angle to set the phase. So [0 1:45] is a 1Hz cycle that starts at 45 degrees:

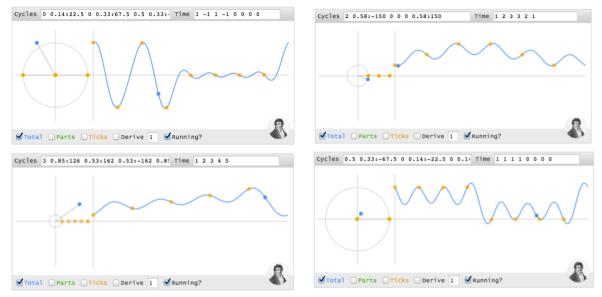


This is a shifted version of [0 1]. On the time side we get [.7 -.7] instead of [1 -1], because our cycle isn't exactly lined up with our measuring intervals, which are still at the halfway point (this could be desired!).

The Fourier Transform finds the set of cycle speeds, strengths and phases to match any time signal.

Our signal becomes an abstract notion that we consider as "observations in the time domain" or "ingredients in the frequency domain".

Enough talk: try it out! In the simulator, type any time or cycle pattern you'd like to see. If it's time points, you'll get a collection of cycles (that combine into a "wave") that matches your desired points.



(http://betterexplained.com/wp-content/uploads/images/time-patterns-20121205-164541.png)

But... doesn't the combined wave have strange values between the yellow time intervals? Sure. But who's to say whether a signal travels in straight lines, or curves, or zips into other dimensions when we aren't measuring it? It behaves exactly as we need at the equally-spaced moments we asked for.

Making A Spike In Time

Can we make a spike in time, like (4 0 0 0), using cycles? I'll use parentheses () for a sequence of time points, and brackets [] for a sequence of cycles.

Although the spike seems boring to us time-dwellers (*one data point, that's it?*), think about the complexity in the cycle world. Our cycle ingredients must start aligned (at the max value, 4) and then "explode outwards", each cycle with partners that cancel it in the future. Every remaining point is zero, which is a tricky balance with multiple cycles running around (we can't just "turn them off").

Let's walk through each time point:

- At time 0, the first instant, every cycle ingredient is at its max. Ignoring the other time points, (4 ? ? ?) can be made from 4 cycles (0Hz 1Hz 2Hz 3Hz), each with a magnitude of 1 and phase of 0 (i.e., 1 + 1 + 1 + 1 = 4).
- At every future point (t = 1, 2, 3), the sum of all cycles must cancel.

Here's the trick: when two cycles are on opposites sides of the circle (North & South, East & West, etc.) their combined position is zero (3 cycles can cancel if they're spread evenly at 0, 120, and 240 degrees).

Imagine a constellation of points moving around the circle. Here's the position of each cycle at every instant:

```
Time 0 1 2 3
-----
0Hz: 0 0 0 0
1Hz: 0 1 2 3
2Hz: 0 2 0 2
3Hz: 0 3 2 1
```

Notice how the the 3Hz cycle starts at 0, gets to position 3, then position "6" (with only 4 positions, 6 modulo (http://betterexplained.com/articles/fun-with-modular-arithmetic/) 4 = 2), then position "9" (9 modulo 4 = 1).

When our cycle is 4 units long, cycle speeds a half-cycle apart (2 units) will either be lined up (difference of 0, 4, 8...) or on opposite sides (difference of 2, 6, 10...).

OK. Let's drill into each time point:

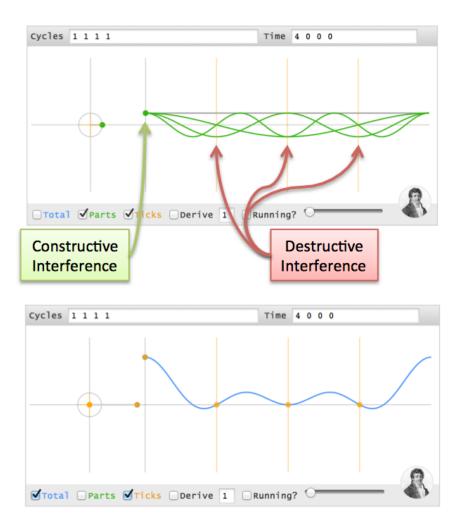
- Time 0: All cycles at their max (total of 4)
- Time 1: 1Hz and 3Hz cancel (positions 1 & 3 are opposites), 0Hz and 2Hz cancel as well. The net is 0.
- Time 2: 0Hz and 2Hz line up at position 0, while 1Hz and 3Hz line up at position 2 (the opposite side). The total is still 0.
- Time 3: 0Hz and 2Hz cancel. 1Hz and 3Hz cancel.
- Time 4 (repeat of t=0): All cycles line up.

The trick is having individual speeds cancel (0Hz vs 2Hz, 1Hz vs 3Hz), or having the lined-up pairs cancel (0Hz + 2Hz vs 1Hz + 3Hz).

When every cycle has equal power and 0 phase, we start aligned and cancel afterwards. (I don't have a nice proof yet -- any takers? -- but you can see it yourself. Try $[1\ 1]$, $[1\ 1\ 1]$, $[1\ 1\ 1]$ and notice the signals we generate: $(2\ 0)$, $(3\ 0\ 0)$, $(4\ 0\ 0\ 0)$).

In my head, I consider these signals "time spikes": they have a burst of activity for a single instant, and are zero otherwise (the fancy name is a <u>delta function</u> (http://en.wikipedia.org/wiki/Dirac delta function).)

Here's how I visualize the initial alignment, followed by a net cancellation:



Moving The Time Spike

Not everything happens at t=0. Can we change our spike to $(0 \ 4 \ 0 \ 0)$?

It seems the cycle ingredients should be similar to $(4\ 0\ 0\ 0)$, but the cycles must align at t=1 (one second in the future). Here's where phase comes in.

Imagine a race with 4 runners. Normal races have everyone lined up at the starting line, the $(4\ 0\ 0\ 0)$ time pattern. Boring.

What if we want everyone to *finish* at the same time? Easy. Just move people forward or backwards by the appropriate distance. Maybe granny can start 2 feet in front of the finish line, Usain Bolt can start 100m back, and they can cross the tape holding hands.

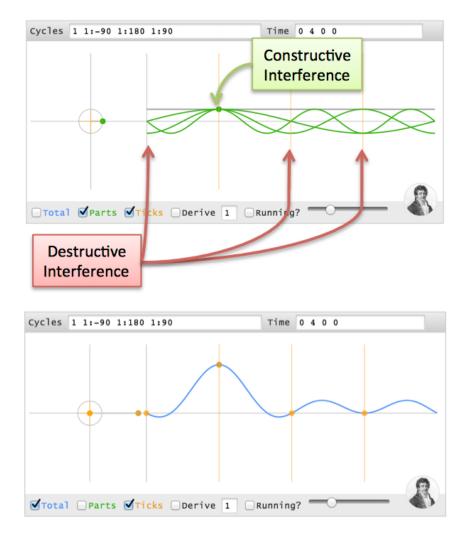
Phase shifts, the starting angle, are delays in the cycle universe. Here's how we adjust the starting position to delay every cycle 1 second:

- A 0Hz cycle doesn't move, so it's already aligned
- A 1Hz cycle goes 1 revolution in the entire 4 seconds, so a 1-second delay is a quarter-turn. Phase shift it 90 degrees backwards (-90) and it gets to phase=0, the max value, at t=1.
- A 2Hz cycle is twice as fast, so give it twice the angle to cover (-180 or 180 phase shift -- it's across the circle, either way).
- A 3Hz cycle is 3x as fast, so give it 3x the distance to move (-270 or +90 phase shift)

If time points $(4\ 0\ 0\ 0)$ are made from cycles $[1\ 1\ 1\ 1]$, then time points $(0\ 4\ 0\ 0)$ are made from $[1\ 1:-90\ 1:180\ 1:90]$. (Note: I'm using "1Hz", but I mean "1 cycle over the entire time period").

Whoa -- we're working out the cycles in our head!

The interference visualization is similar, except the alignment is at t=1.



Test your intuition: Can you make $(0\ 0\ 4\ 0)$, i.e. a 2-second delay? 0Hz has no phase. 1Hz has 180 degrees, 2Hz has 360 (aka 0), and 3Hz has 540 (aka 180), so it's $[1\ 1:180\ 1\ 1:180]$.

Discovering The Full Transform

The big insight: our signal is just a bunch of time spikes! If we merge the recipes for each time spike, we should get the recipe for the full signal.

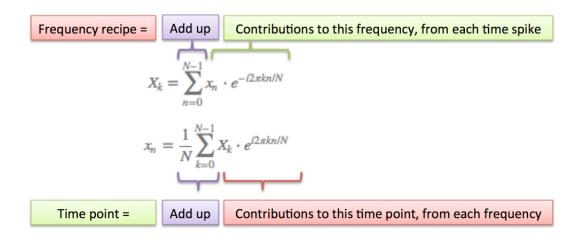
The Fourier Transform builds the recipe frequency-by-frequency:

- Separate the full signal (a b c d) into "time spikes": (a 0 0 0) (0 b 0 0) (0 0 c 0) (0 0 0 d)
- For any frequency (like 2Hz), the *tentative* recipe is "a/4 + b/4 + c/4 + d/4" (the strength of each spike is split among all frequencies)
- Wait! We need to offset each spike with a phase delay (the angle for a "1 second delay" depends on the frequency).
- Actual recipe for a frequency = a/4 (no offset) + b/4 (1 second offset) + c/4 (2 second

offset) + d/4 (3 second offset).

We can then loop through every frequency to get the full transform.

Here's the conversion from "math English" to full math:



A few notes:

- N = number of time samples we have
- n = current sample we're considering (0 .. N-1)
- x_n = value of the signal at time n
- k = current frequency we're considering (0 Hertz up to N-1 Hertz)
- X_k = amount of frequency k in the signal (amplitude and phase, a complex number)
- The 1/N factor is usually moved to the *reverse transform* (going from frequencies back to time). This <u>is allowed</u>

 (http://math.stackexchange.com/questions/58163/dft-why-are-the-definitions-for-inverse-and-forward-commonly-switched), though I prefer 1/N in the forward transform since it gives the *actual* sizes for the time spikes. You can get wild and even use 1/sqrt(N) on both transforms (going forward and back still has the 1/N factor).
- n/N is the percent of the time we've gone through. 2 * pi * k is our speed in radians / sec. e^-ix is our backwards-moving circular path. The combination is how far we've moved, for this speed and time.
- The raw equations for the Fourier Transform just say "add the complex numbers".
 Many programming languages cannot handle complex numbers directly, so you convert everything to rectangular coordinates and add those.

Onward

This was my most challenging article yet. The Fourier Transform has several flavors (discrete/continuous/finite/infinite), covers deep math (Dirac delta functions), and it's easy to get lost in details. I was constantly bumping into the edge of my knowledge.

But there's always simple analogies out there -- I refuse to think otherwise. Whether it's a smoothie or Usain Bolt & Granny crossing the finish line, take a simple understanding and refine it. The analogy is flawed, and that's ok: it's a raft to use, and leave behind once we cross the river.

I realized how feeble my own understanding was when I couldn't work out the transform of $(1\ 0\ 0\ 0)$ in my head. For me, it was like saying I knew addition but, gee whiz, I'm not sure what "1+1+1+1" would be. Why not? Shouldn't we have an intuition for the simplest of operations?

That discomfort led me around the web to build my intuition. In addition to the references in the article, I'd like to thank:

- <u>Scott Young (http://www.scotthyoung.com/blog/)</u>, for the initial impetus for this post
- <u>Shaheen Gandhi (http://shaheengandhi.com)</u>, Roger Cheng, and <u>Brit Cruise (http://britcruise.com/)</u> for kicking around ideas & refining the analogy
- <u>Steve Lehar (http://cns-alumni.bu.edu/~slehar/fourier/fourier.html)</u> for great examples of the Fourier Transform on images
- <u>Charan Langton (http://www.complextoreal.com/chapters/fft1.pdf)</u> for her detailed walkthrough
- <u>Julius Smith (https://ccrma.stanford.edu/~jos/log/)</u> for a fantastic walkthrough of the Discrete Fourier Transform (what we covered today)
- <u>Bret Victor (http://worrydream.com/#!/LearnableProgramming)</u> for his techniques on visualizing learning

Today's goal was to *experience* the Fourier Transform. We'll save the advanced analysis for next time.

Happy math.

Appendix: Projecting Onto Cycles

Stuart Riffle has a great interpretation

(https://web.archive.org/web/20120418231513/http://www.altdevblogaday.com/2011/05/1 the-fourier-transform/) of the Fourier Transform:

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

To find the energy at a particular frequency, spin your signal around a circle at that frequency, and average a bunch of points along that path.

Imagine spinning your signal in a centrifuge and checking for a bias. I have a correction: we must spin *backwards* (the exponent in the equation above needs a negative sign). You already know why: we need a phase *delay* so spikes appear in the *future*.

Appendix: Another Awesome Visualization

<u>Lucas Vieira (http://toxicdump.org/blog/)</u>, author of excellent <u>Wikipedia animations</u> (http://en.wikipedia.org/wiki/User:Kieff/Gallery), was <u>inspired</u> (http://blog.matthen.com/post/42112703604/the-smooth-motion-of-rotating-circles-can-be-used) to make this interactive animation:

Fourier Toy - Click to Open (http://toxicdump.org/stuff/FourierToy.swf)



(http://toxicdump.org/stuff/FourierToy.swf)

(Detailed list

(http://www.reddit.com/r/math/comments/17v8cv/visualized_fourier/c89c3pa) of control options)

The Fourier Transform is about cycles added to cycles added to cycles. Try making a "time spike" by setting a strength of 1 for every component (press Enter after inputting each number). Fun fact: with enough terms, you can draw any shape, even Homer Simpson (http://www.youtube.com/watch?v=OVuU2YCwHjw&feature=youtu.be).

Ptolemy and Homer (Simpson)







Appendix: Using the code

All the code and examples are open source (MIT licensed, do what you like).

- <u>Interactive example (http://betterexplained.com/examples/fourier/)</u> (view source)
- Github gist (https://gist.github.com/129d477ddb1c8025c9ac)
- <u>Reddit discussion</u>
 (http://www.reddit.com/r/programming/comments/156a9i/an_interactive_guide_to_context=3) on details of the computation, I'm pb_zeppelin

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Questions & Contributions

Translation <u>Chinese Translation (http://select.yeeyan.org/view/478578/406382)</u>

select.yeeyan.org

Resource Analogy: The Fourier Transform "probes" for which sine/cosine is

present. (http://www.earlevel.com/main/2002/08/31/a-gentle-

introduction-to-the-fft/) www.earlevel.com

Resource	Fourier tutorial using R, incorporates the interactive diagrams above (http://www.di.fc.ul.pt/~jpn/r/fourier/fourier.html) www.di.fc.ul.pt
Resource	Nice interactive diagrams on many elements of signals & Fourier Transform (http://jackschaedler.github.io/circles-sines-signals/) jackschaedler.github.io
Resource	<u>Great video on mechanical device to do Fourier Analysis</u> (https://www.youtube.com/watch?v=NAsM30MAHLg) www.youtube.com

<u>Ask a Question (http://betterexplained.com/articles/an-interactive-guide-to-the-fourier-transform/#tab-faq)</u>

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Ask Qu	uestion		
Ask Qu	uestion		

Continue Discussion 324 replies



<u>niko</u>

Those animations are AMAZING. Mad mad props. But I don't really get how the time points are spaced... "The time points are spaced at the fastest frequency. A 1Hz signal needs 2 time points for a start and stop (a single data point doesn't have a frequency). The time values [1 -1] shows the amplitude at these equally-spaced intervals."

Like for a 1 Hz signal why are you measuring at 2 points, for a 2 Hz signal at 3 points, for a 3 Hz signal at 4 points and so on? Does this have something to do with the Nyquist-Shannon sampling theorem?

1 reply



kalid Founder

Thanks Waldir, just fixed.





Anonymous_User

FINALLY THIS MAKES SENSE



<u>gdiehl</u>

A couple examples of using the Fourier integrals and series on a known signal (e.g. a simple one like 1+sin(t)) and showing how the 'parts' are extracted would be helpful.

I tried and failed to get what I expected (e.g. 1 for the DC component for this litle signal above at w=0) when I attempted to evaluate the integral form, which probably just means I forgot how to integrate properly (but Maxima blew up on it too).

Kalid - As always, thanks *so much* for your efforts! I look forward to you tackling the Laplace transform one of these days:)



<u>amit</u>

The analogy is rather out-of-place, confusing and seems like an unnecessary detour. Most important of all, the author seems to lack primary insight on the subject himself. Smoothie is a whole and its ingredients are parts. It is not the same with signals at all. They are all whole in their own domain. The analogy is totally misplaced and the information provided only partly correct. The key idea of FT - change of variable - is not emphasized at all. Only well informed people should be allowed to author such articles. That said, there are a couple of good insights for new learners - 1. decomposition of a signal into other signals, though that hardly warrants such a confusing and misplaced analogy and 2. representation of complex numbers on the z-plane, though this may not be the best place for such a long discussion on that.

<u>1 reply</u> <u>Feb '13</u>



satheesh

Hi Genius,

Thanks for the article ,It was awesome like always but i have a doubt. What does negative values in time mean like -1 the graph is always positive in x and y value is just projection of point in x so i can't get negative meaning of time



daniel

I think I can answer that question also. When the frequency domain is discrete, it makes sense to talk about non-infinite amounts of each component, as you describe. The trouble is, when the frequency domain is continuous, it makes less sense to talk about the amount at each frequency, and a "amount density" is more applicable. After all, in the real world there is not really such thing as a pure sine tone; everything has some width in frequency space. But if a finite width has noninfinitesimal "amounts" at every frequency, then the total amount is infinite, which makes no sense. This is the same principle why discussing the mass of an object in detail requires a notion of a local mass density, because if every POINT in the object had its own noninfinitesimal mass, the whole thing would have infinite mass. The Dirac delta is a formal way to revive the concept of a pure sine frequency even after we have evolved to the notion of densities. The Dirac delta is the representation of a "point mass" in the "density" way of thinking about things.

I think what you said about the height of the transform being equal to the amplitude is correct in the discrete frequency interpretation, because there densities DON'T make sense, while "amounts" do.



Anonymous_User

Is there anyway to visualize orthogonal signals?Or understand it intuitively instead of just saying that their dot product is zero. I'm studying trigonometric fourier series and having a hard time grasping how each multiple of fundamental frequency component acts the same way like orthogonal vectors that when adding them together they doesn't interfere and you can add frequency components with right amplitudes to the sum and you don't need to correct already calculated amplitudes when adding a new term.



<u>000</u>

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<u>pete</u>

Great explanation for something that is so seemingly complicated!



steve

Worth the wait - only 22 years for me!



<u>vijay</u>

Keep it coming! Thanks!

It must have been hard work. And fun in putting in a lot of imagination to get this done. Great work.



glenndiehl

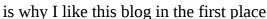
Stephen,

Thanks for replying. I actually did recognize that e^0 is one. The integral becomes integral $(1+\sin(t)dt)$, which evaluates to t-cos(t). But, evaluating that from -inf to +inf blows up, which is no big surprise.

Dividing by 't', as you mentioned, doesn't quite get me there either.

The formula in the first bullet of section 2.4 in http://www.civilized.com/files/newfourier.pdf (see Gary Knott's post above - thanks Gary) does work (divide by 1/p (the fundmental frequency) and adjust the limits of integration to be one cycle), but I don't follow (yet, anyway, but I'm not giving up) how we can just replace the +/-infinities in the formal definition of the integral.

At least I finally got the answer I was looking for, but needless to say I would have failed the exam (which





Glenn



frankabbing

I love the smoothie metaphor!

Feb '13



kalid Founder

Hah, just a curious learner here. Negative values in time, for the signal you mean? In this case, if it's an infinitely-repeating signal (that's what the Fourier Transform assumes), then -1 means "1 second before the current cycle started, i.e. near the end of the previous cycle". Sort of like "-1AM" might mean 11PM of the night before.

Feb '13



kalid Founder

Hi Jose, thanks -- that's a really neat interpretation. Yep, it's almost like a "dot product" where you are seeing how big and overlap you can get (and different phases will have different overlaps).

Dec '12



Anonymous_User

Hey, man, this is really great. I like to use DFTs for stochastic processes and look at their underlying trend (this is similar to harmonic regression, which is kinda' cool). Thanks for this!

Feb '13



Anonymous_User

Wonderful!



stephenrbesch

Daniel.

Don't get me wrong. It is not at all that I disagreed with your assertions, especially since they are formally correct in every way. They are also really cool and insightful and your clarification is very adequate,. The confusion centers arount the traditional use of the delta function in the time domain as an infinitely high pulse of zero duration whose integral =1. Also, invoking this interpretation involves understanding a lot of really difficult math involving distributions. etc. I am also a little preplexed by the use of "s", usually reserved for the complex variable of integration of the LaPlace transform. My only intent is to help clarify the workings of the Fourier concept using simple, approachable math.



stephenrbesch

"But if a finite width has noninfinitesimal "amounts" at every frequency, then the total amount is infinite, which makes no sense". The problem with this statement is that the width is finite and therefore there are only a finite number of finite amounts and the sum is therefore finite! If I recall my history, this is precisely the polemic that raged for years regarging whether in the limit, the differential becomes vanishingly small on the one hand or 0 on the other.. The point was that as the slice becomes smaller and smaller it's contribution also becomes smaller, and in the limit, the sum converges on the integral. The argument is whether an infinite number of infintely small pieces totals up to a finite value. Certainly we know that when we integrate a simple function that the answer almost always gives us the area bounded by the curve, not an infinite value, and in this regard, for all practical considerations it does not seem to matter if we consider that the differential truly reaches 0 in the limit or not.I suppose that we could argue about this for years, just as mathematicians did in the 19th century, However, it appears that Paul Dirac must have finally resolved this debate with the invention of the Delta Function. In the interest of all the other readers of this posting, I think we should put the whole matter to rest. At least that's what I am going to do!



<u>daniel</u>

Yes, what you are saying is correct, but the "amounts" have to tend toward zero as the "pieces" become smaller to give a finite result. I am just explaining why setting the amount at each piece to a constant (non infinitesimal) would give trouble.



daniel

The reason the Fourier transform of 1+sin(t) seems to be blowing up is because it does... The Fourier transform of a pure Fourier mode will always just be a delta function centered around the appropriate frequency. In the case of the zero frequency component, we expect zero anywhere away from zero, but an infinitely thin spike around zero.



stephenrbesch

Glenn.

I should have mentioned the 1 cycle issue. Here's the explanation you are looking for (I hope). Since Sin and Cos are periodic, the integral over one cycle is exactly the same as the integral over an infinite number of cycles. Also, since the integral 1+sin(t)dt evaluates to t-cos(t), then divide by t to get 1-cos(t)/t. In the limit of infinity, this clearly becomes exactly equal to 1 - not to mention that cos(t) can never exceed one anyway, so this integral can never blow up - at least not with real valued t. If t is complex, I'm not sure what may happen. Why Maxima can't cope with this probably has something to do with how it deals with the definition of infinity. Also, if I remember the deatils, the 1/p issue has to do with changing the limits of the integration to cover exactly one cycle. It is essentially the same as integrating from -2Pi to +2Pi. You might also consider forgetting about the negative frequency part of the spectrum. For all real valued data (that is, all real data!) it is exactly the mirror image of the positive spectrum anyway,



stephenrbesch

Daniel,

Are you sure about this? I've done a lot of fourier analysis on single frequency sine waves with and without a DC offset and indeed, the result is always a spike at the fundamental frequency of the sine wave and if there is a DC offset, another at the origin. However, these are not of infinite amplitude - quite the contrary, the heights are always exactly equal to the amplitude of the sine wave and it's DC component. If this were not to be the case, the Fourier transform would not really be very useful for AC signal analysis. I suspect that you are thinking about some other interesting property of the transform and I would like to have you clarify what you are thinking about - perhaps the transform of the delta function?



<u>000</u>

Math is as easy as pie! but needs someone to bake it as good as you did! Great!!

Congra!





daniel

I'd be happy to explain myself. I say the Fourier Transform of $\$1+\sin(x)$ has three spikes, one at \$s=0,\$ and one at both $\$s=-\frac{1}{2\pi}$ and $\$s=\frac{1}{2\pi}$ respectively. Namely, for $\$f(x)=1+\sin(x)$:

 $\frac{1}{2i}\left[\left(s-\frac{1}{2\pi}\right)-\left(s+\frac{1}{2\pi}\right)\right].$

If you don't buy this, well, let's just check it. According to the definition given at the beginning of the post of the Inverse Fourier Transform, we have

 $$\Big\{-\inf y^{\left(s+\frac{1}{2i}\left(s-\frac{1}{2\pi}\right)-\det(s+\frac{1}{2\pi}\right)\right\} e^{-2\pi i s x} \$

Then when we perform the integration, the delta functions yield the integrand evaluated at the value of s that makes the argument of that delta function zero. This gives

 $\$e^0+\frac{1}{2i}\left[e^{ix}-e^{-ix}\right]=\boxed{1+\sin(x)},$ \$\$ as desired.

Feb '13



<u>angel</u>

The old Yamaha DX synthesizers series used frequency modulation to create, from 4 to 6 sine waveforms, quite complex sounds.

Jun '13



Anonymous_User

I didn't enjoy this article as much as the others. This is because the recipe analogy didn't feel intuitive. If anyone is having difficulties processing this, I highly recommend reading James' http://practicalcryptography.com/miscellaneous/machine-learning/intuitive-guide-discrete-fourier-transform/. It's way more verbose and more intuitive (if you understand simple statistic concepts like correlation and variance).

<u>Jun '13</u>



<u>elpetersen</u>

This is awesome! Nicely intuitive.



whippy

@Matija: re Q4, I'm not sure if this helps, but the above examples/annotations are all based on an overall time interval of 1 second, and 1Hz, 2Hz, etc. Probably in a real application the overall time interval would not be 1 second, and therefore the frequencies would change accordingly. For example (as I understand it), if the time interval was 1/10th second then the waves would have frequencies of 0Hz, 10Hz, 20Hz, etc, for as many samples as you have in that interval. If the time interval were 2 seconds then you would actually have 0Hz, 0.5Hz, 1Hz, 1.5Hz, etc, which might be of some use in your case. Also bear in mind the article uses the unit "Hz" (and "seconds") a little loosely in places, which it explains. Hope this helps.



<u>claude</u>

I had asked Kalid about the Fourier Transform a while back and he had emailed a great brief explanation. I just returned to this site to see whether there were any updates and I'm happy to see that the Fourier Transform is on here. This is a fantastic website!

Feb '13



kalid Founder



Thanks Claude!



maarten

"One of my giant confusions was separating circles from sinusoids."

This was one of my most notable Aha!-moments during the Analysis class as well. Glad to see it explained clearly here. You might find the website below very interesting, especially 'phasor phactory' (in case you haven't encountered it yet).

http://www.jhu.edu/signals/index.html

Keep up this useful work,

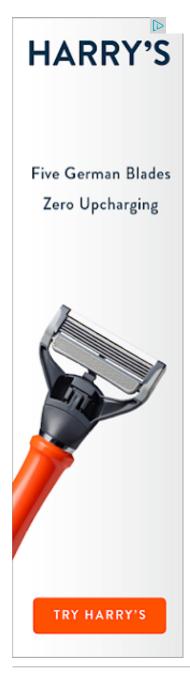
M.

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