

Fourier Transform

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Definition

Let $f : \mathbb{R} \rightarrow \mathbb{R}$ be a given real-valued function. **Fourier Transform** of function $f(t)$ is defined by

$$\hat{f}(\omega) = \int_{-\infty}^{\infty} f(t) e^{-i2\pi\omega t} dt.$$

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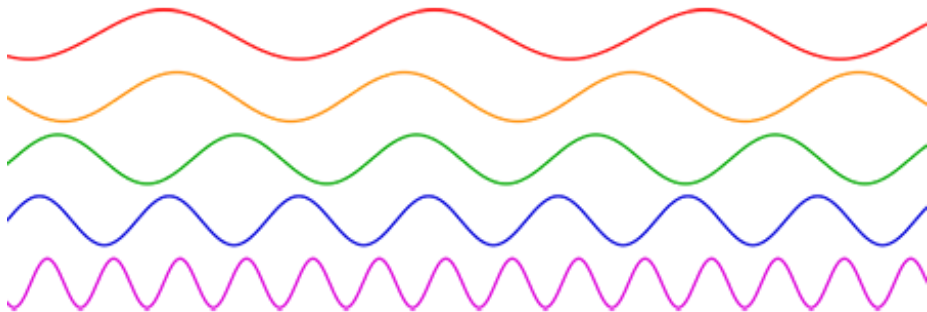
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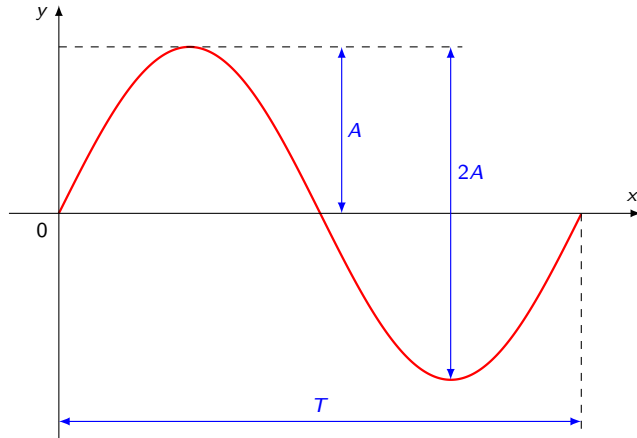
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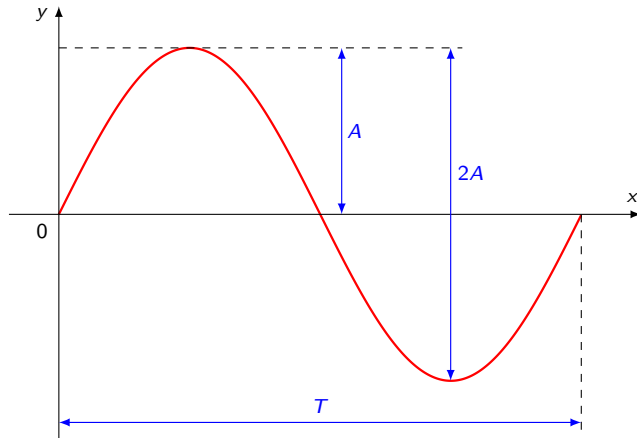
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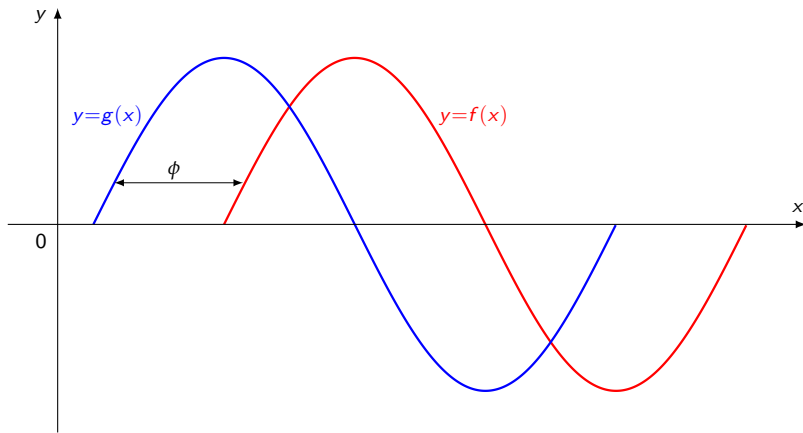
We are saying that Fourier Transform moves us from **time domain** to **frequency domain**.





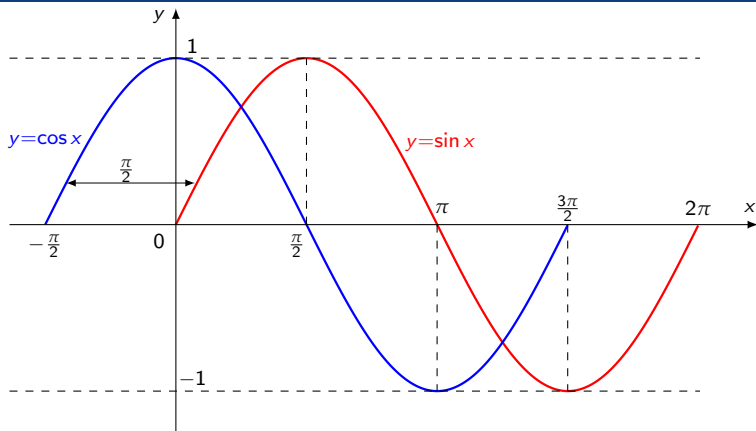


$y = A \sin(kx),$ Amplitude $A,$ Period $T = \frac{2\pi}{k}$

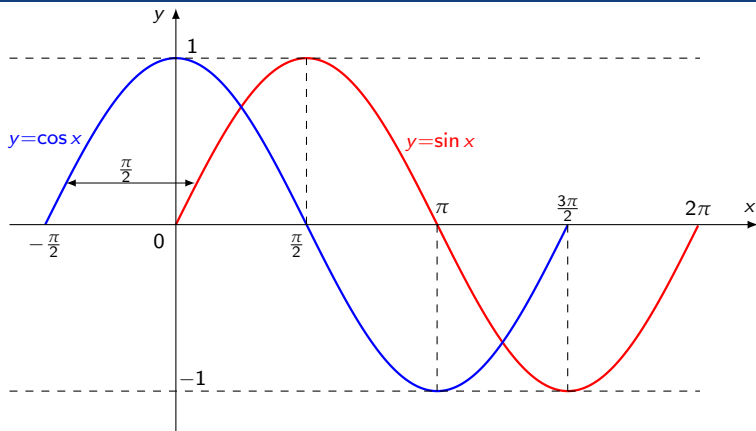


$$f(x + \phi) = g(x), \quad g(x - \phi) = f(x)$$

$\sin x$ and $\cos x$ functions



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$$\sin\left(x + \frac{\pi}{2}\right) = \cos(x), \quad \cos\left(x - \frac{\pi}{2}\right) = \sin(x)$$

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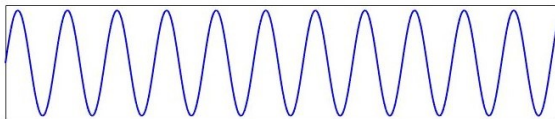
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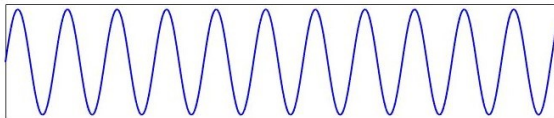
We should try to filter out the noise and find the underlying signal properties, like periodicity, frequency and so on.

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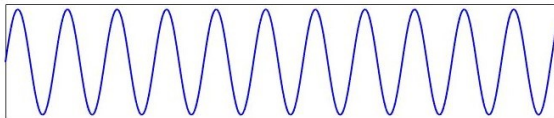


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The information that signal carries is the frequency of that wave.

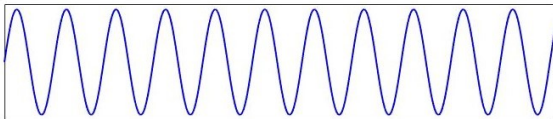
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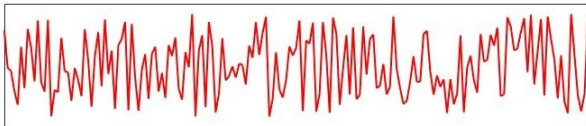
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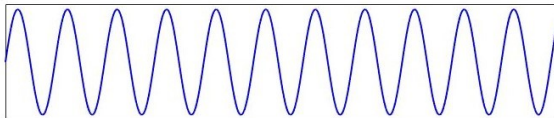


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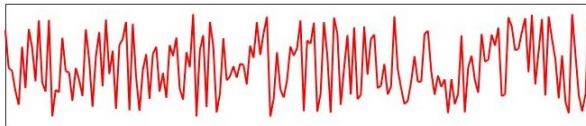


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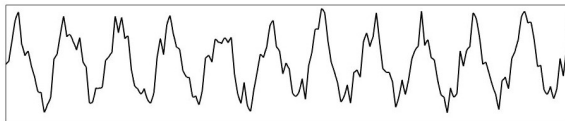
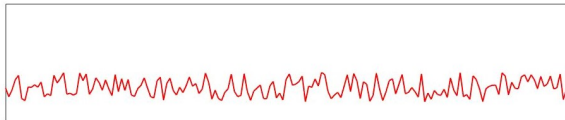
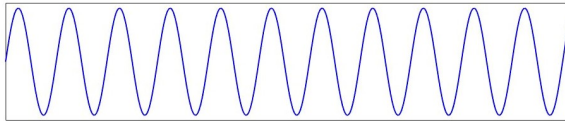


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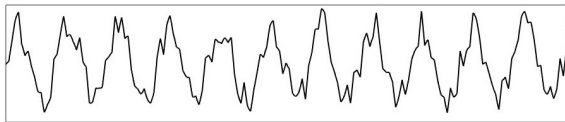
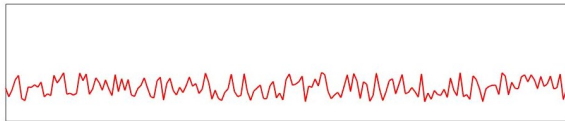
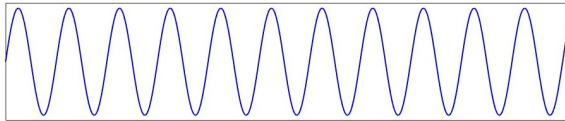
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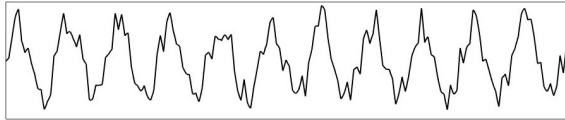
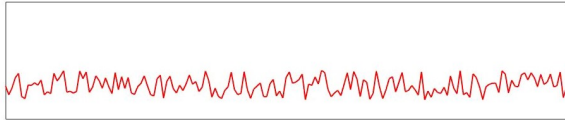
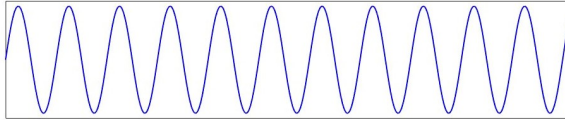
If we increase the strength of the noise, when do we lose the information contained in the original signal?



Signal and noise with amplitude ratio 4:1

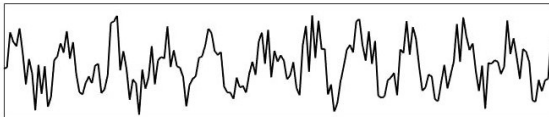
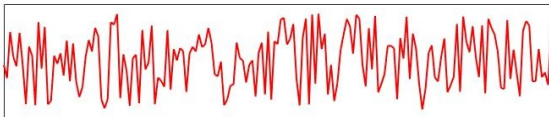
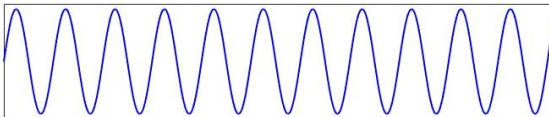


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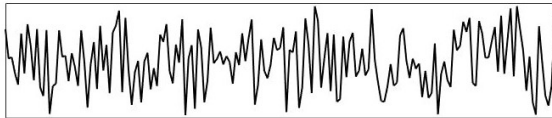
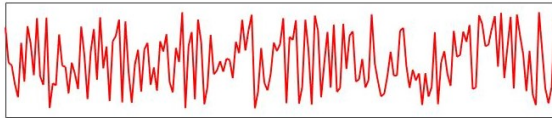
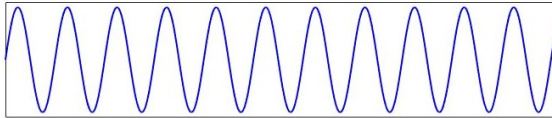
However, it is still recognizable as a noisy wave with the same frequency as the original signal.
The information has not yet been lost!



The combined **signal** + **noise** is now very noisy.

It appears that we are losing information from our original signal.

Signal and noise with amplitude ratio 1:2.5



The combined **signal** + **noise** is now as random as the noise.

It appears that we definitely lost any information from our original signal.

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Suppose, signal $S(t)$ is a pure sine function, where t is time measured in seconds. Our test probe is the function:

$$P(t) = \sin(2\pi\omega t)$$

whose frequency is $\frac{2\pi\omega}{2\pi} = \omega$ cycles per second.

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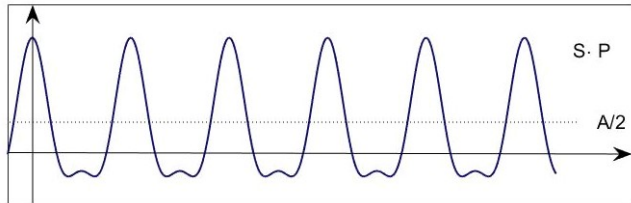
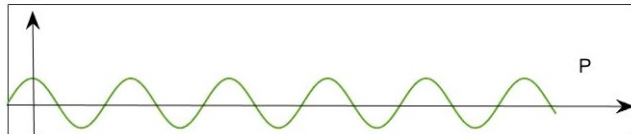
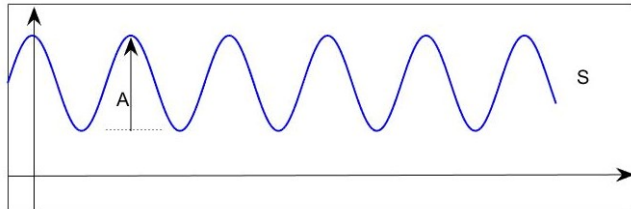
What is happening to

$$S(t) \cdot P(t)?$$

Suppose first that clean signal
 $S(t)$ has **the same**
frequency as probe $P(t)$.

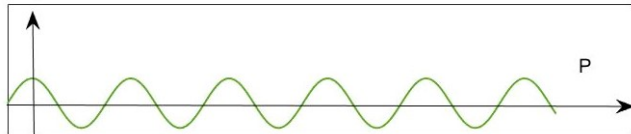
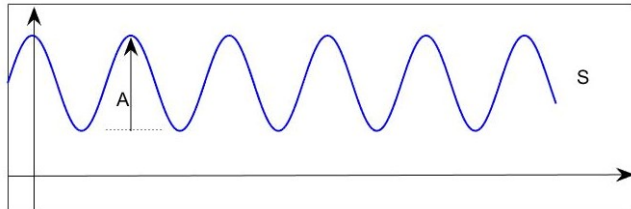
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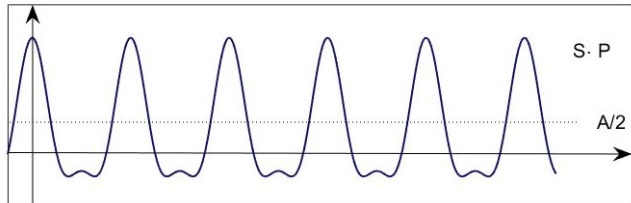


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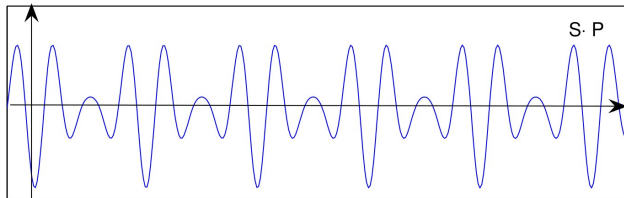
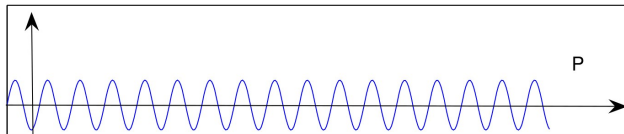
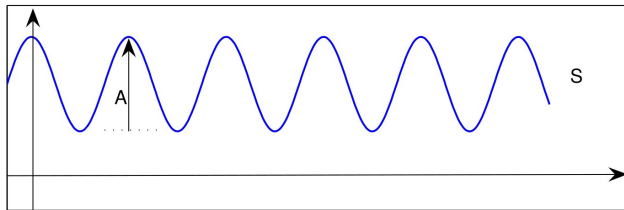
The average value of the product $S(t) \cdot P(t)$ is $\frac{A}{2}$



Now, suppose that a clean signal $S(t)$ has a **different frequency** than probe $P(t)$.

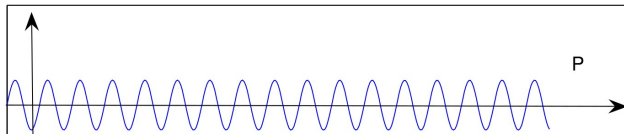
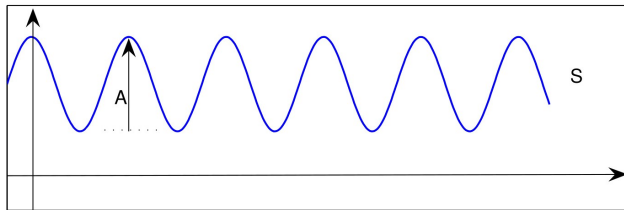
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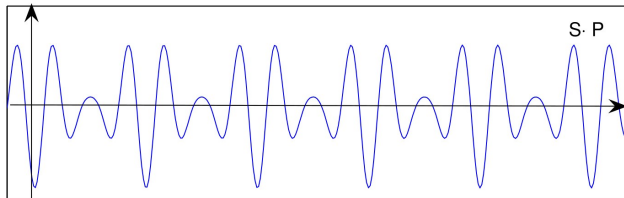


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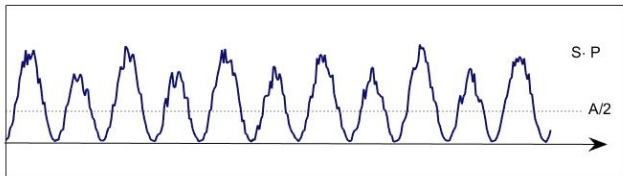
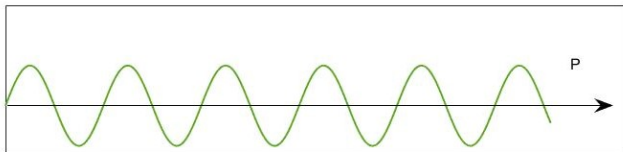
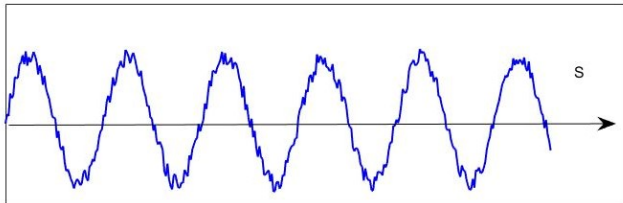
The average value of the product $S(t) \cdot P(t)$ is 0



Signal $S(t)$ plus some noise
with amplitude ratio 4 : 1.

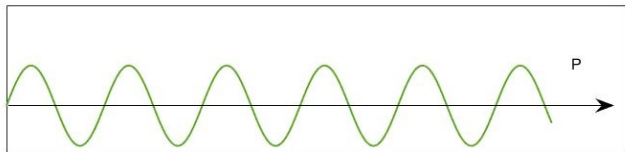
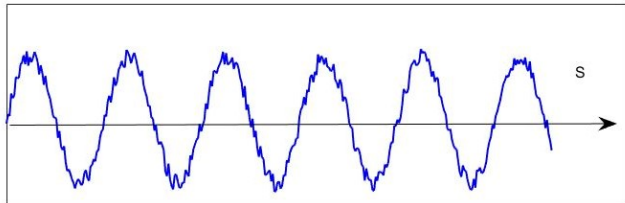
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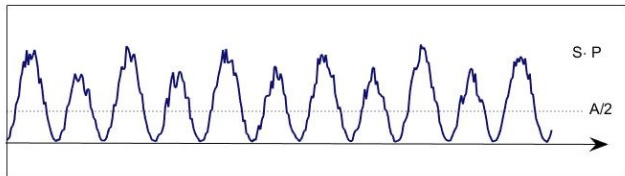


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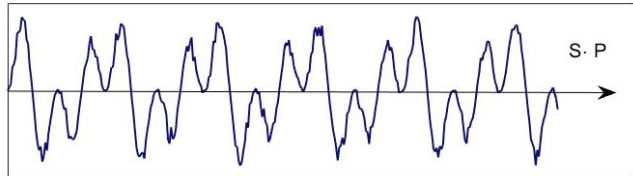
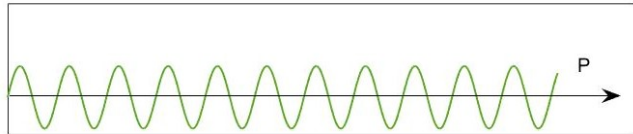
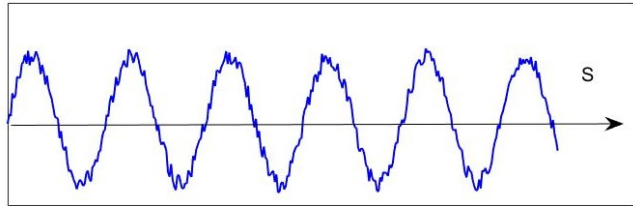
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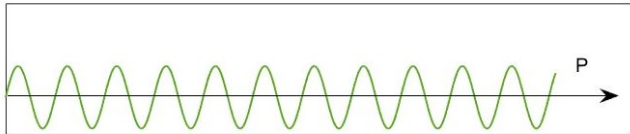
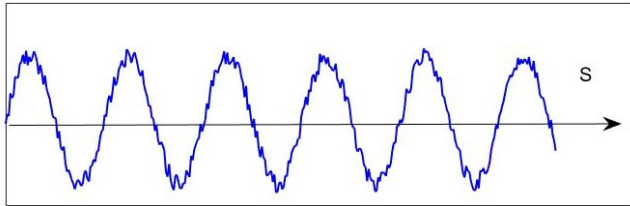
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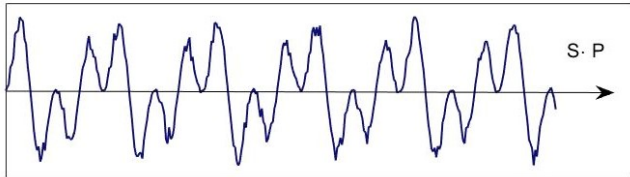


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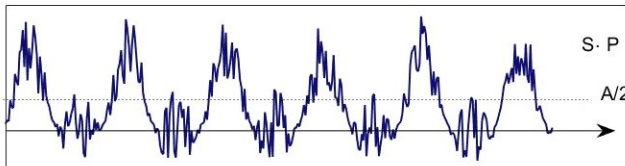
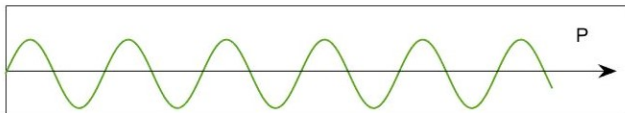
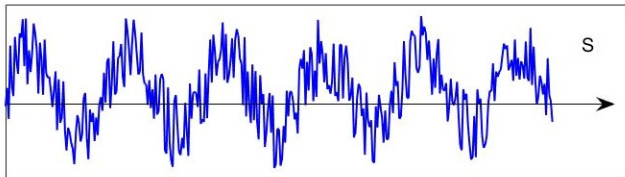
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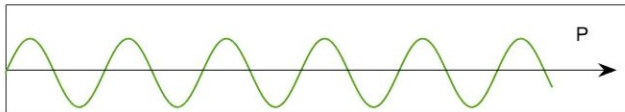
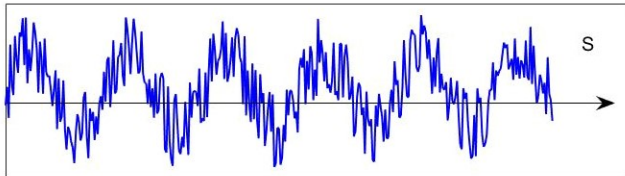
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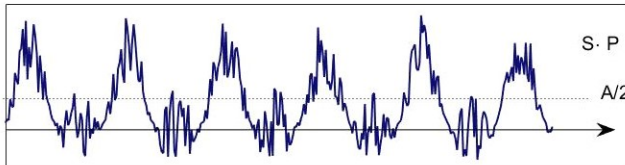


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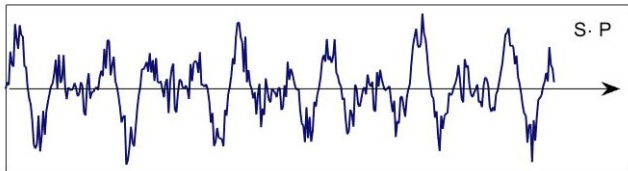
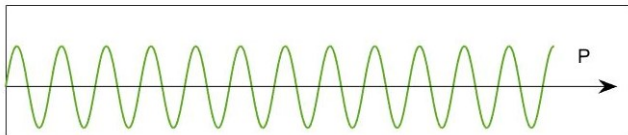
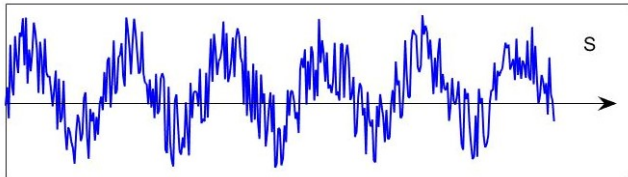
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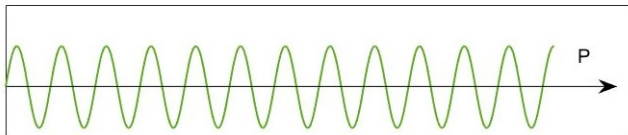
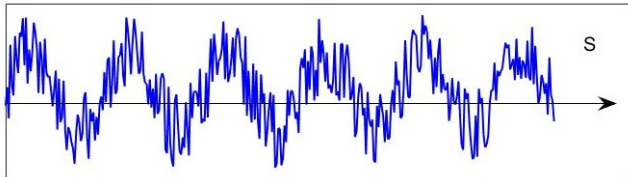
Signal does not match the test frequency

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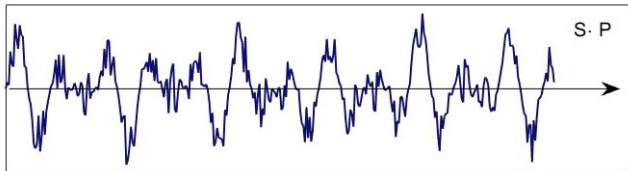


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Signal $S(t)$ plus some noise
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The average value of the
product $S(t) \cdot P(t)$ is 0



Recall the following definition:

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Given a function $f : [a, b] \rightarrow \mathbb{R}$, the average value of function f on interval $[a, b]$ is

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Our detector is the average value of the product of the signal $S(t)$ and the test probe $P(t) = \sin(2\pi\omega t)$:

$$\text{Frequency detector } D(\omega) = \frac{1}{b-a} \int_a^b S(t) \sin(2\pi\omega t) dt$$

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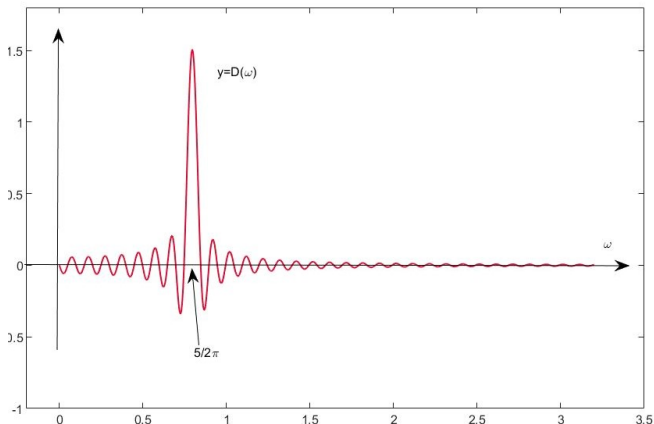
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Integrating with respect to t , we get

$$D(\omega) = \frac{3}{10(4\pi^2\omega^2 - 25)} \left(5 \cos(50) \sin(20\pi\omega) - 2\pi\omega \sin(50) \cos(2\pi\omega) \right).$$

Plot the graph of function $D(\omega)$ (on next page)

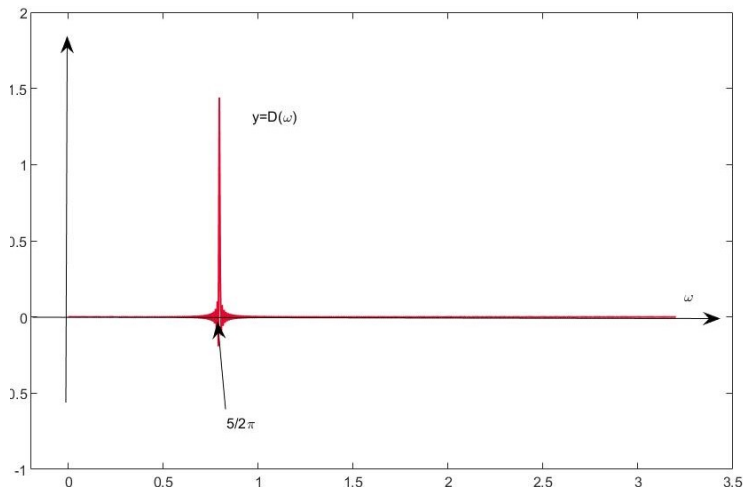
Example 1



For most frequencies the value of detector $D(\omega)$ is close to 0.
There is a single strong peak, which you can find at $\omega \approx 0.795$ cycles/s.
As it happens the frequency of signal $S(t) = 3 \sin(5t)$ is $\frac{5}{2\pi} \approx 0.795$.
Moreover the height of the peak is $1.5 = \frac{3}{2}$, half of the amplitude.

Example 1

We can repeat this test for $t \in [0, 60]$ and note that $D(\omega)$ has a sharper peak.



For most signals $D(\omega)$ is computed numerically for a sequence of values of parameter ω .

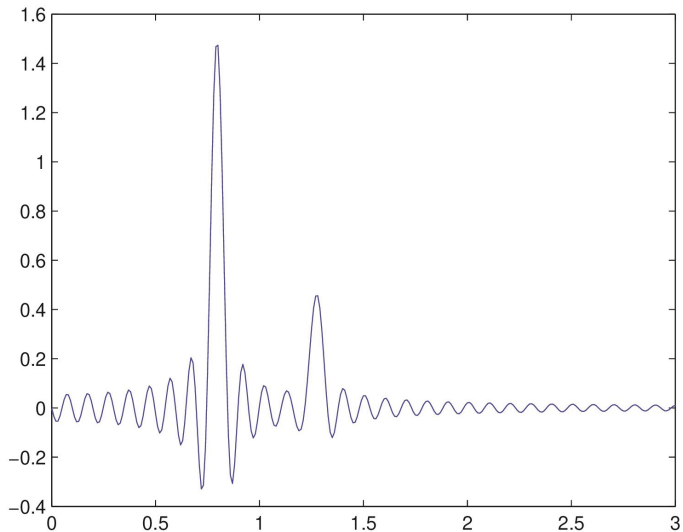
Example 2



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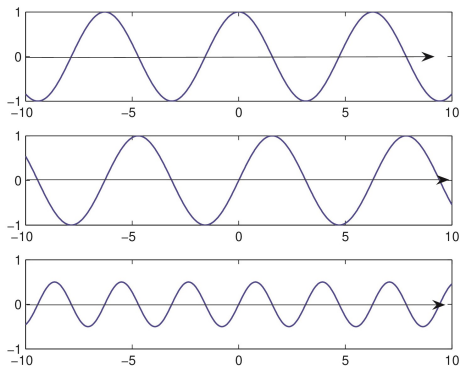
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However, something quite different will happen if we change the signal $S(t) = \cos t$ instead of $S(t) = \sin t$.

The product $S(t) \cdot P(t)$ (with the probe having the same frequency as the signal) will oscillate around 0 and will have the average value 0.



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$$\begin{aligned}\sin(bt - \varphi) &= \cos \varphi \sin(bt) - \sin \varphi \cos(bt) \\ &= M \sin(bt) + N \cos(bt).\end{aligned}$$

Therefore, the sine probe $P_S(t) = \sin(2\pi\omega t)$ will detect $M \sin(bt)$ and the cosine probe $P_C(t) = \cos(2\pi\omega t)$ will detect $N \cos(bt)$.

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In this case, the height of the detector at a peak will equal the amplitude of the signal at that frequency.

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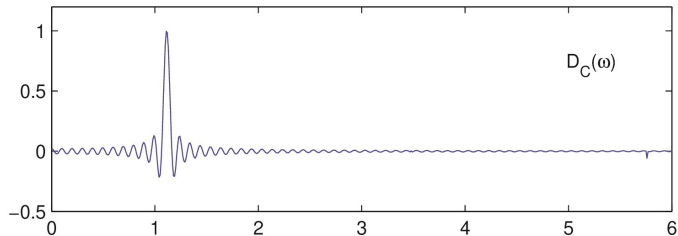
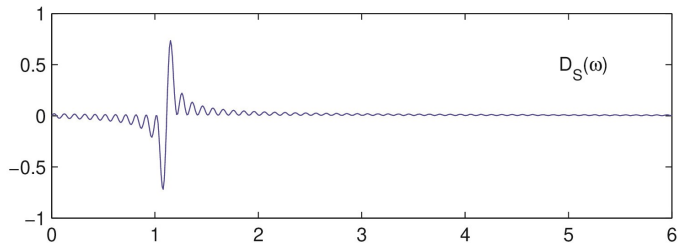
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Fourier transforms and their inverses are used in photo restoration, in the enhancement of the digitized pictures sent back from cameras in space and in filtering the audio signal in modern stereo systems.

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Resonance occurs all around us.

Sine and cosine detectors can be combined to determine the only the strength of the different frequencies that occur in the signal:

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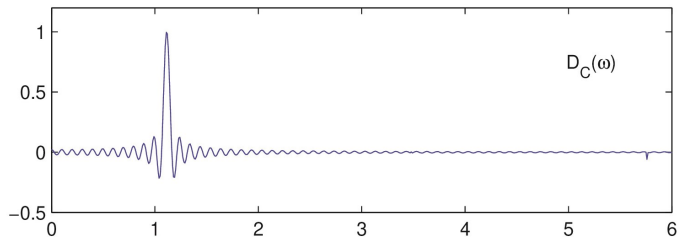
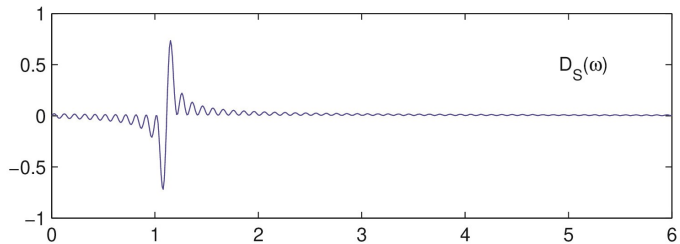
If the phase φ is nonzero, then

$$D_S\left(\frac{7}{2\pi}\right) = A \cos \varphi, \quad D_C\left(\frac{7}{2\pi}\right) = A \sin \varphi$$

and

$$D_P\left(\frac{7}{2\pi}\right) = \sqrt{\left(D_S\left(\frac{7}{2\pi}\right)\right)^2 + \left(D_C\left(\frac{7}{2\pi}\right)\right)^2} = \sqrt{(A \cos \varphi)^2 + (A \sin \varphi)^2} = A.$$

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```
function [DDS,DDC,DDP]=detector2(omega,a,b)
    DS = @(c)(quad(@(x) (myfun(x).*sin(2*pi*c.*x)), a, b));
    DC = @(c)(quad(@(x) (myfun(x).*cos(2*pi*c.*x)), a, b));
    for i = 1 : length(omega)
        DDS(i)=2/(b-a)*DS(omega(i));
        DDC(i)=2/(b-a)*DC(omega(i));
    end
    DDP=sqrt(DDS.^2+DDC.^2);
    function y=myfun(x)
        y=cos(4*pi*x);
    end
end
```

To see how power spectrum detects the frequencies consider two signals:

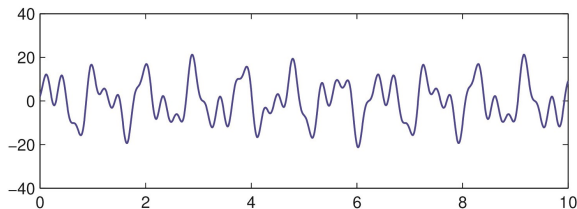
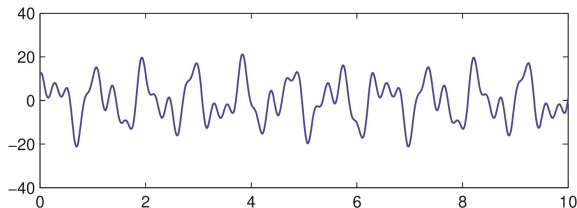
$$S_1(t) = 10 \sin(7t) + 7 \cos(13t) + 5 \cos(23t)$$

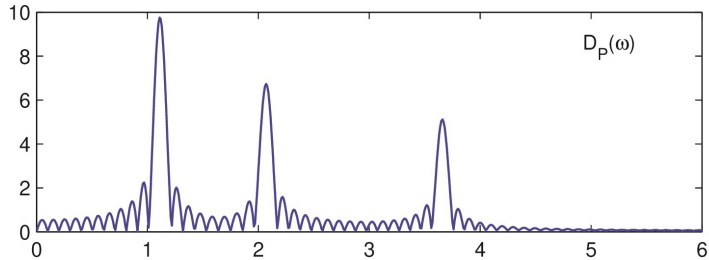
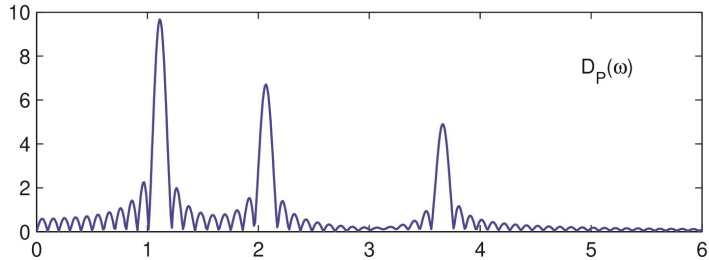
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The same power spectrum will be obtained also for any signal of the form

$$S_1(t) = 10 \sin(7t - \varphi_1) + 7 \cos(13t - \varphi_2) + 5 \cos(23t - \varphi_3)$$

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Now, we return to the signal+noise problem that we raised at the beginning.

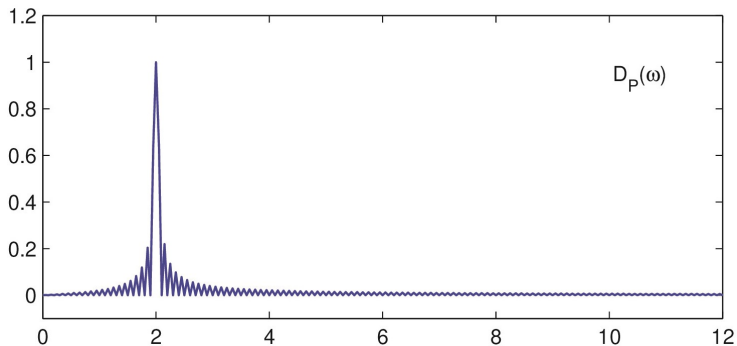
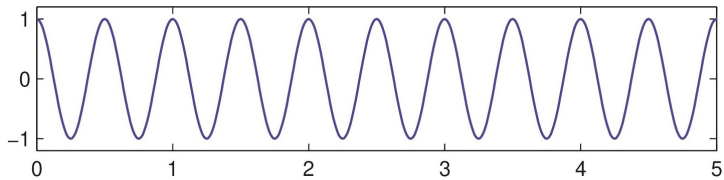
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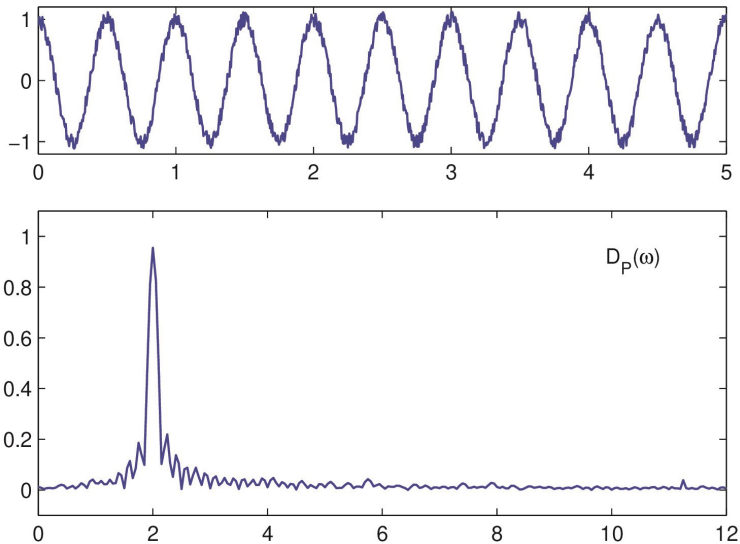
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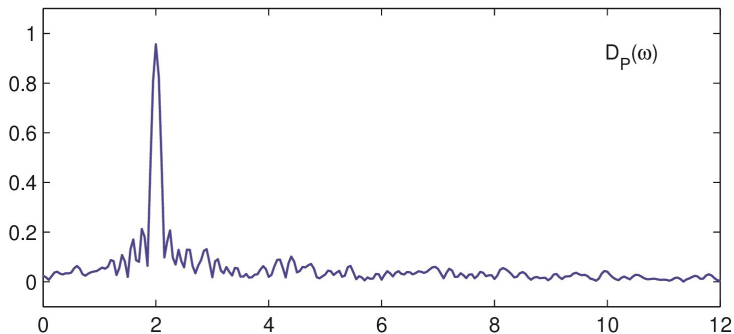
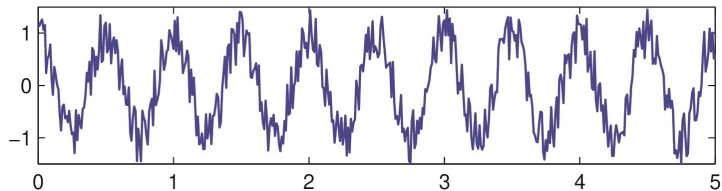
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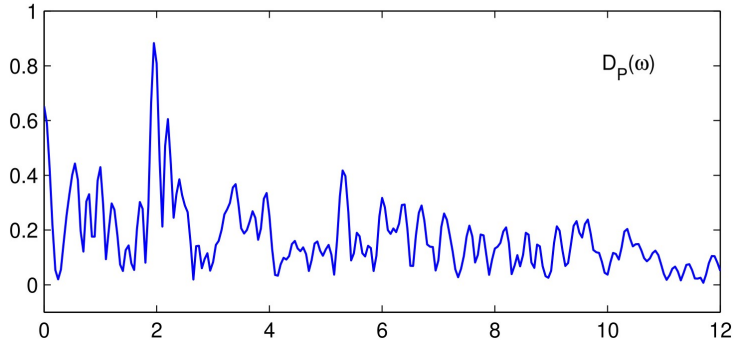
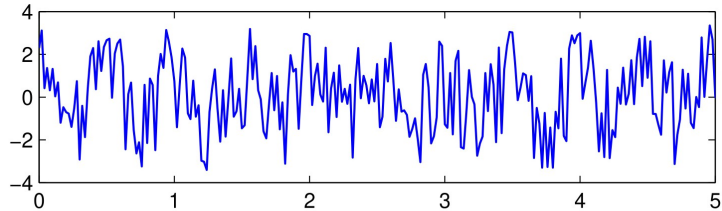
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For simplicity we will take the frequency of the pure signal to be 2 cycles per second.









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Let $f : \mathbb{R} \rightarrow \mathbb{R}$ be a given real-valued function. **Fourier Transform** of function $f(t)$ is defined by

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We are saying that Fourier Transform moves us from **time domain** to **frequency domain**.

Recall that

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Definition

$$\begin{aligned}\hat{f}(\omega) &= \int_{-\infty}^{\infty} f(t) e^{-i2\pi\omega t} dt \\ &= \int_{-\infty}^{\infty} f(t) \cos(2\pi\omega t) dt - i \int_{-\infty}^{\infty} f(t) \sin(2\pi\omega t) dt.\end{aligned}$$

Recall that

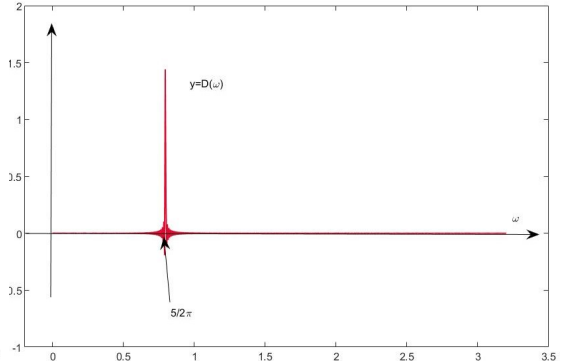
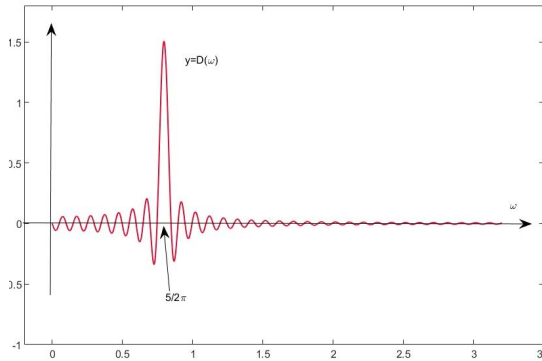
$$e^{i\alpha x} = \cos(\alpha x) + i \sin(\alpha x).$$

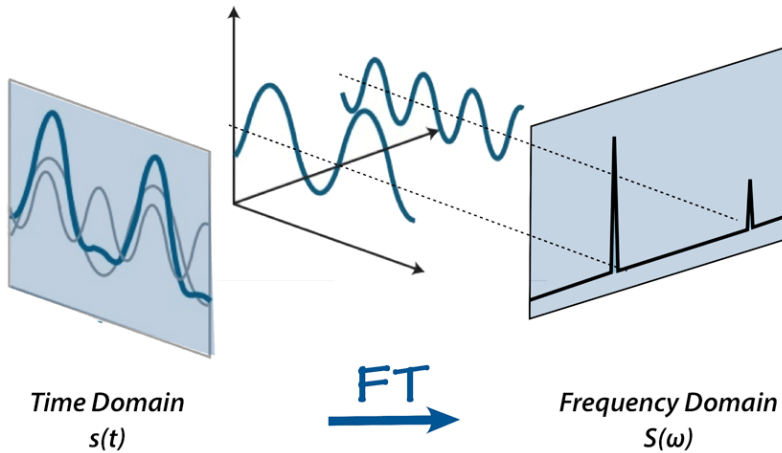
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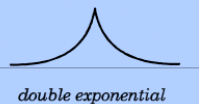
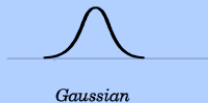
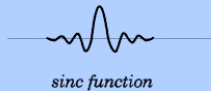
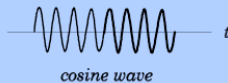
Thus, Cosine detector (Almost Cosine Fourier Transform) is the Real part of the Fourier Transform, whether Sine detector (Almost sine Fourier Transform) is the Imaginary part, respectively.

$$f(t) = 3 \sin(5t)$$





Signal $s(t)$



Fourier Transform $S(\omega)$

