Corso di Laurea in Fisica – UNITS ISTITUZIONI DI FISICA PER IL SISTEMA TERRA

### LINEAR SYSTEMS

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**Green's function** (GF) is a basic solution to a linear differential equation, a building block that can be used to construct many useful solutions.

If one considers a linear differential equation written as:

L(x)u(x)=f(x)

where L(x) is a linear, self-adjoint differential operator, u(x) is the unknown function, and f(x) is a known nonhomogeneous term, the GF is a solution of:

$$L(x)u(x,s)=\delta(x-s)$$

$$G(x,s)$$



### Why GF is important?



If such a function G can be found for the operator L, then if we multiply the second equation for the Green's function by f(s), and then perform an integration in the s variable, we obtain:

$$\int L(x)G(x,s)f(s)ds = \int \delta(x-s)f(s)ds = f(x) = Lu(x)$$
$$L\int G(x,s)f(s)ds = Lu(x)$$

$$u(x) = \int G(x,s)f(s)ds$$

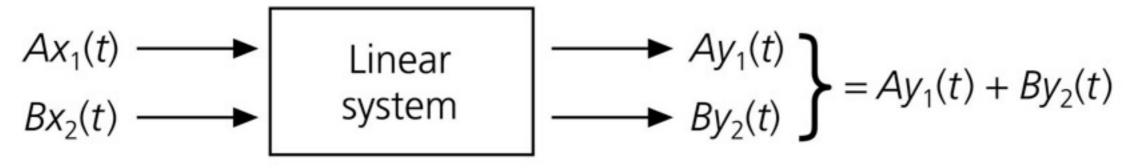
Thus, we can obtain the function u(x) through knowledge of the Green's function, and the source term. This process has resulted from the linearity of the operator L.







Figure 6.3-1: Definition of a linear system.



x(t) = 
$$\int x(\tau)\delta(\tau - t)d\tau$$
  
 $\int x(\tau)h(\tau - t)d\tau$   
x(t) \* h(t) = y(t)

(remember GF definition)









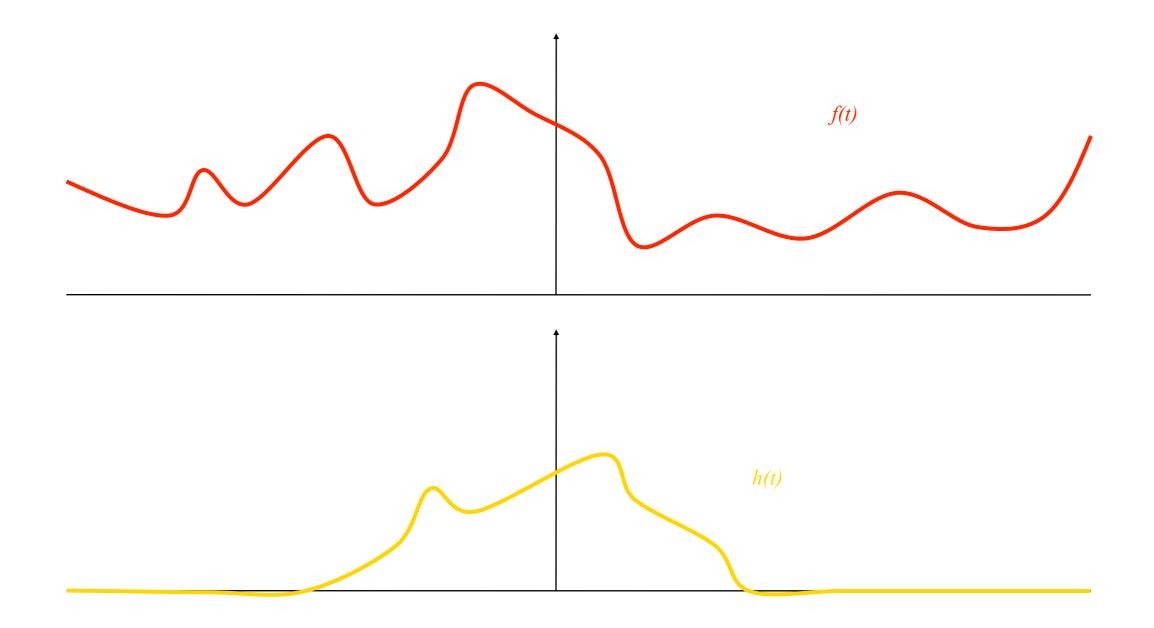
# $f(t) * h(t) = \int_{-\infty}^{\infty} f(\tau)h(t - \tau)d\tau$







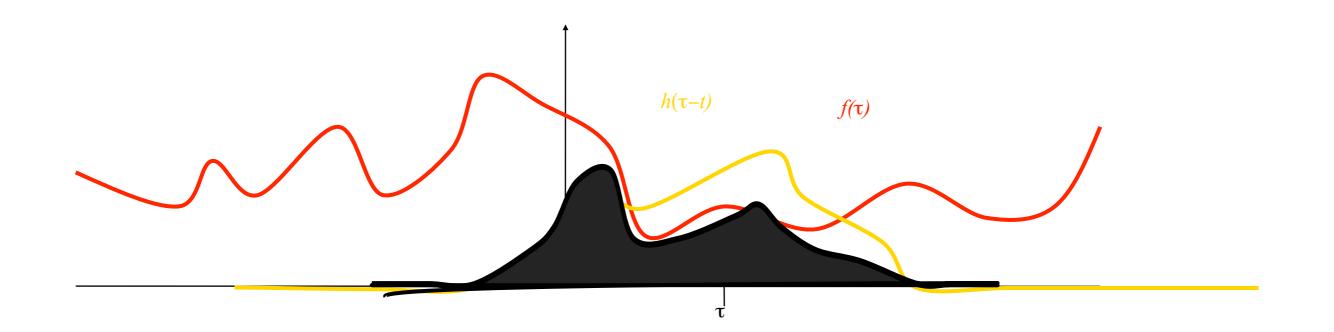












Fabio Romanelli

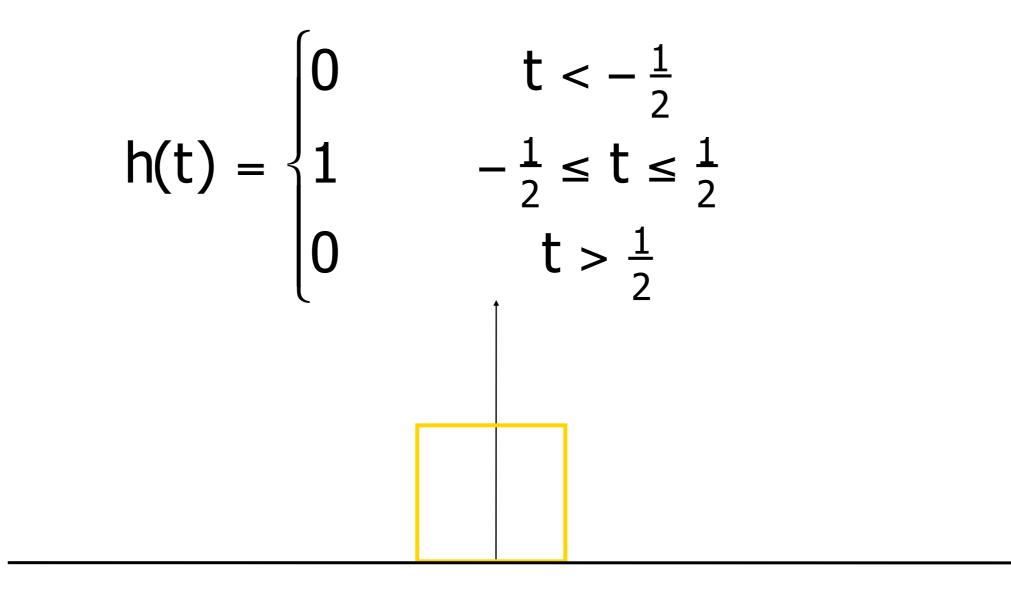
Linear Systems







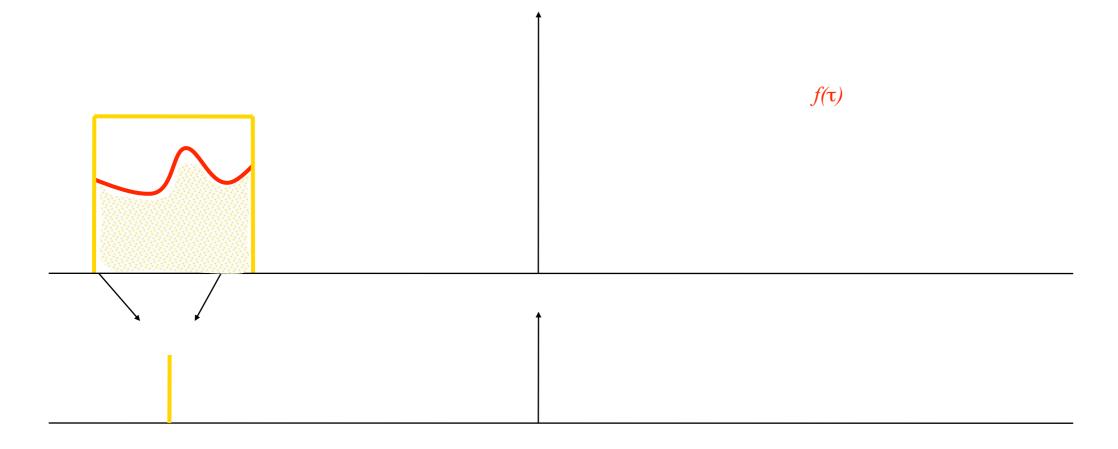
### Consider the function (box filter):





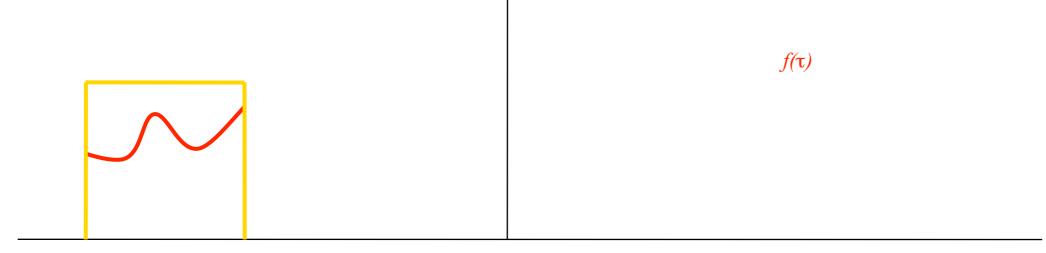








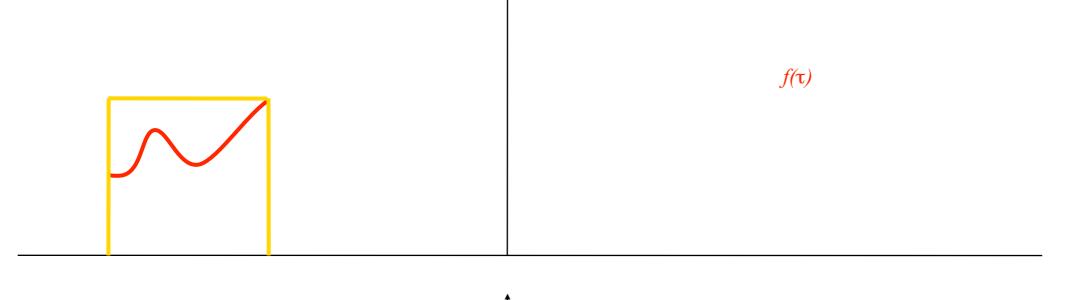






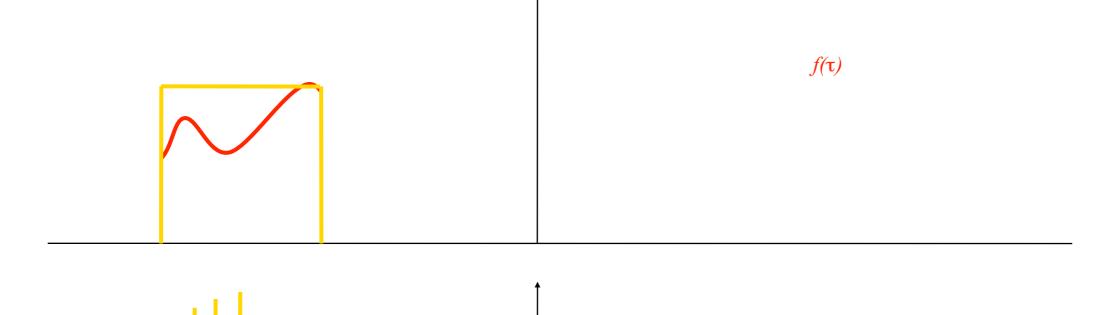






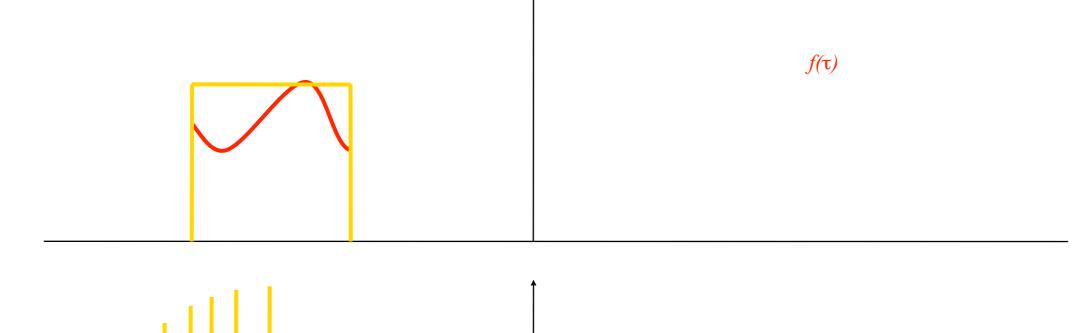






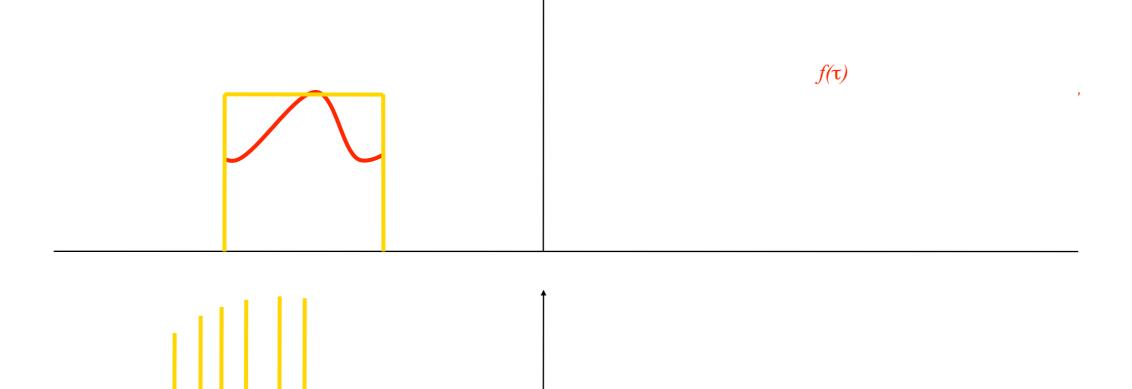






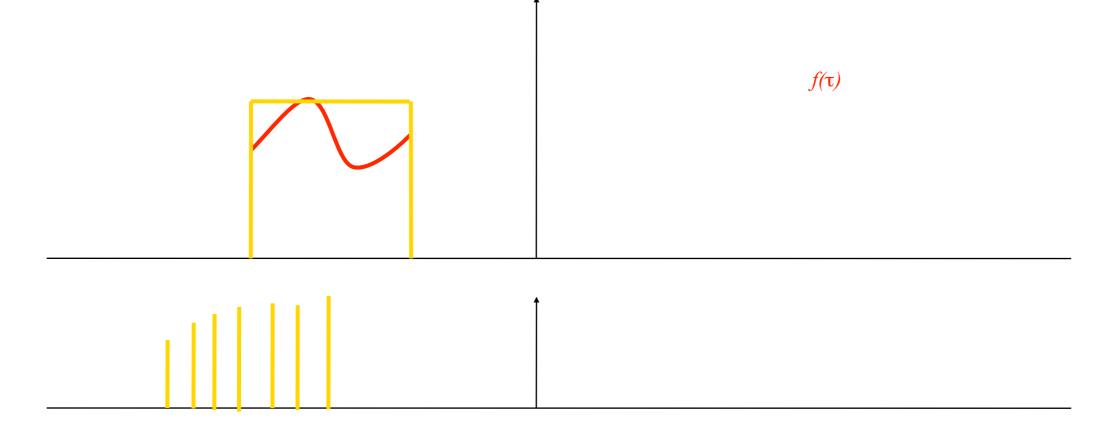






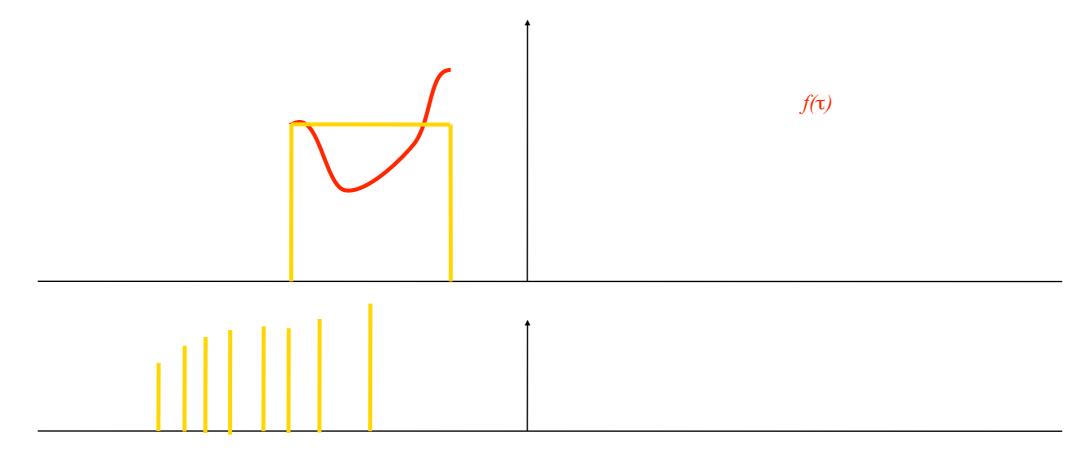






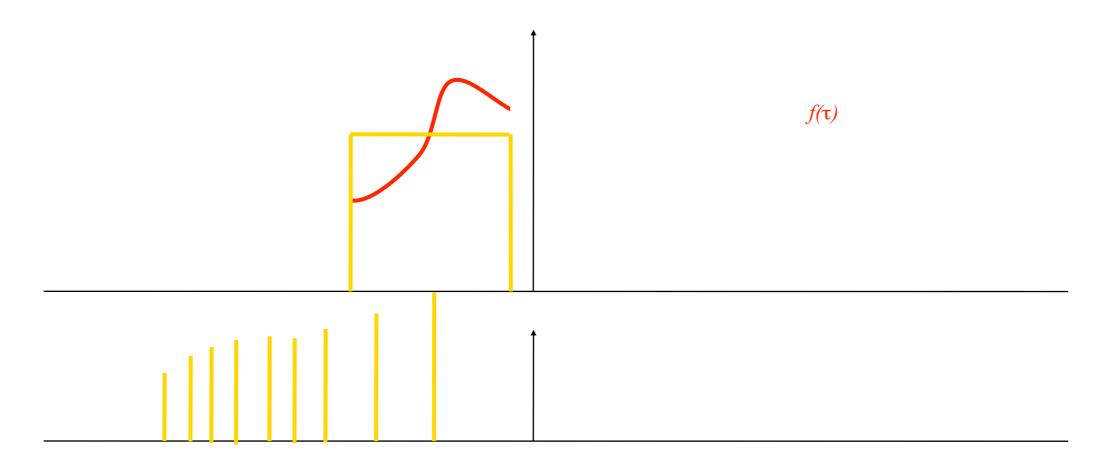






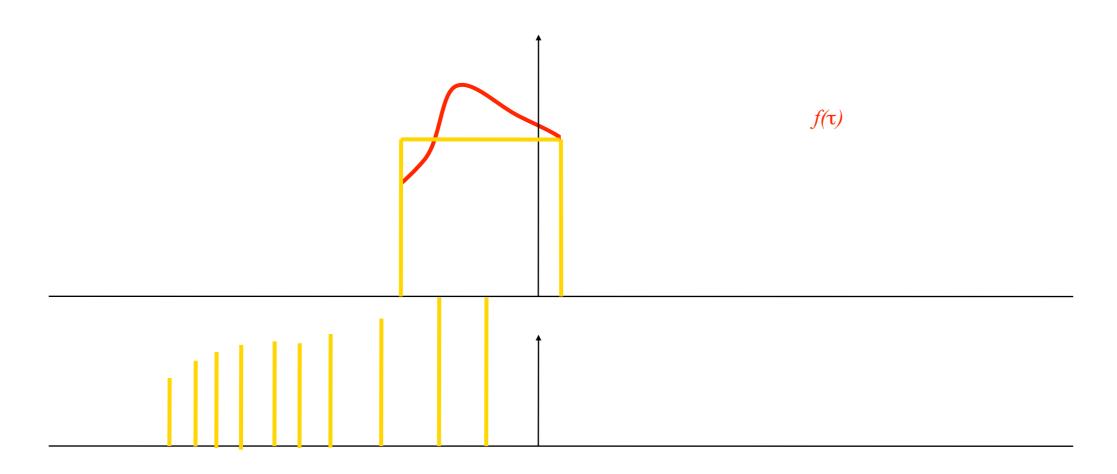






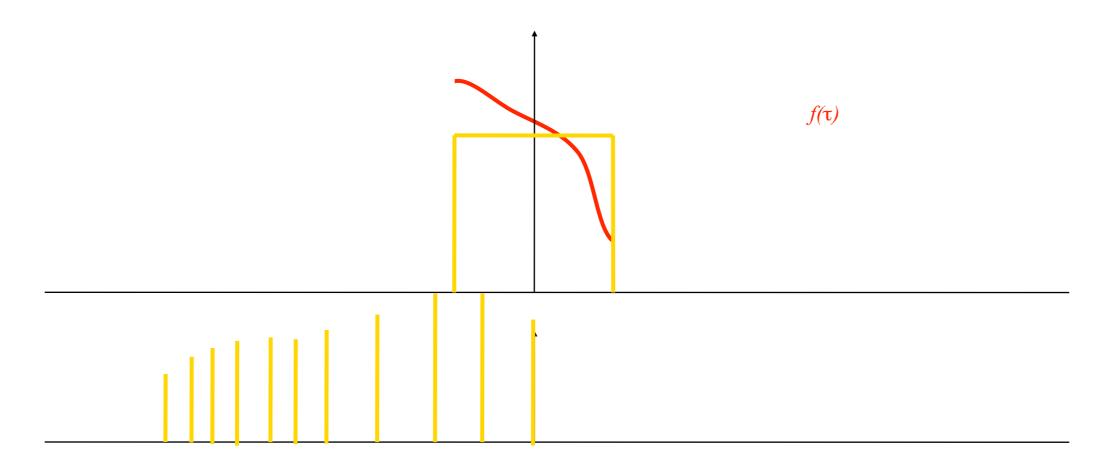






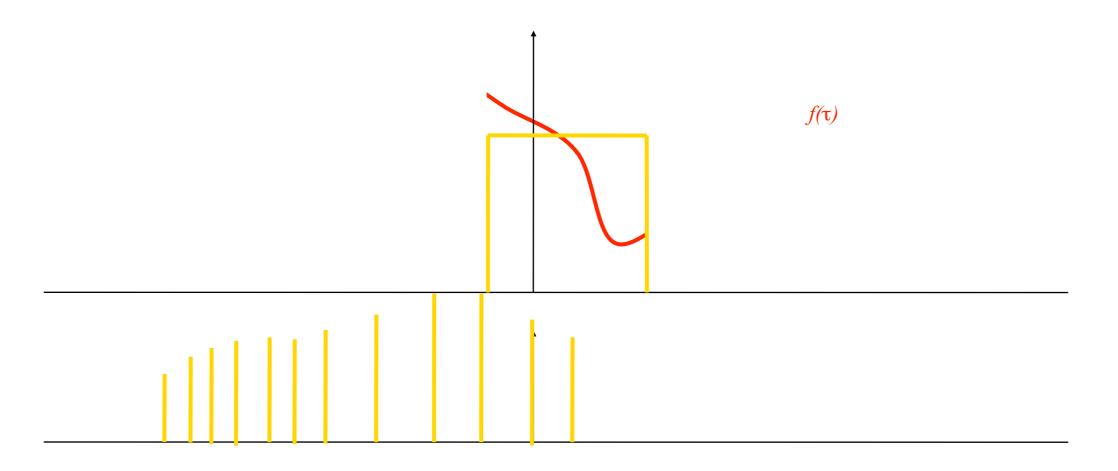






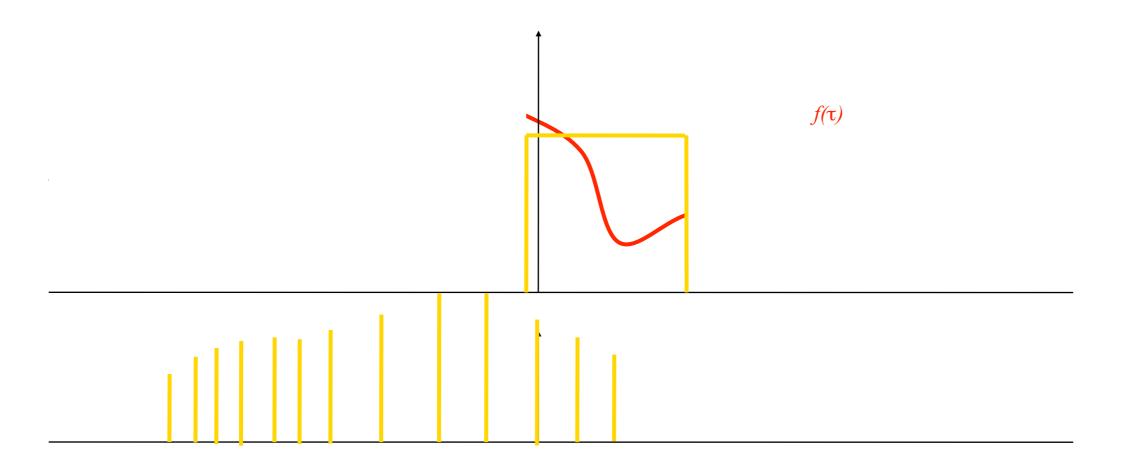






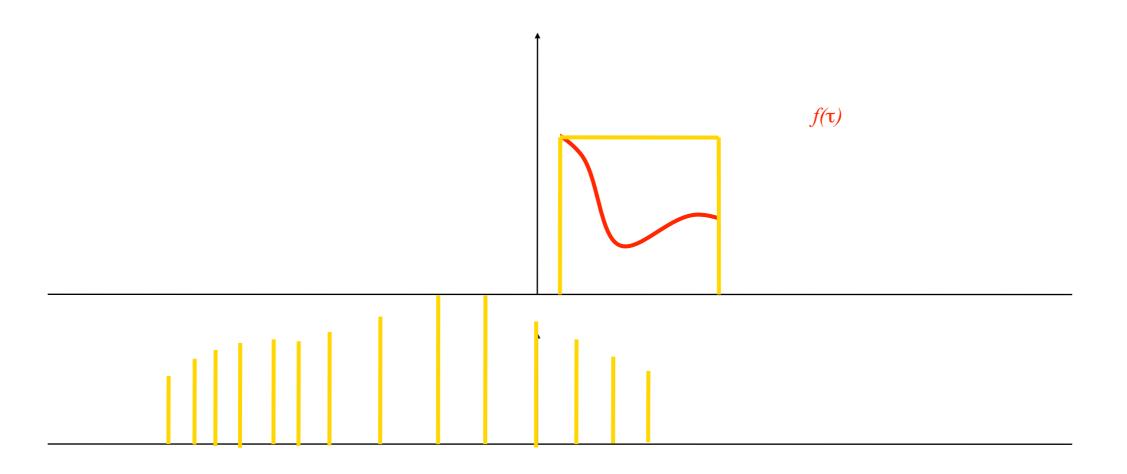






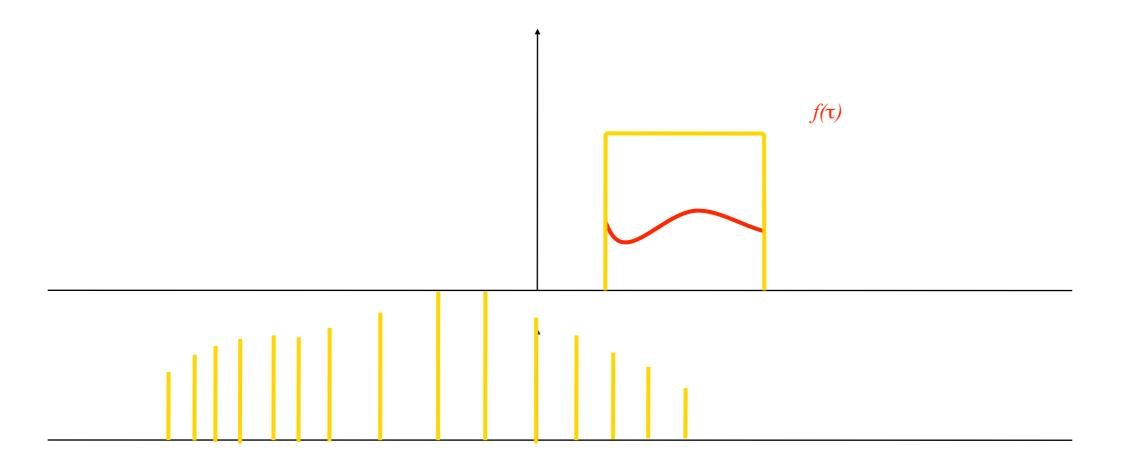






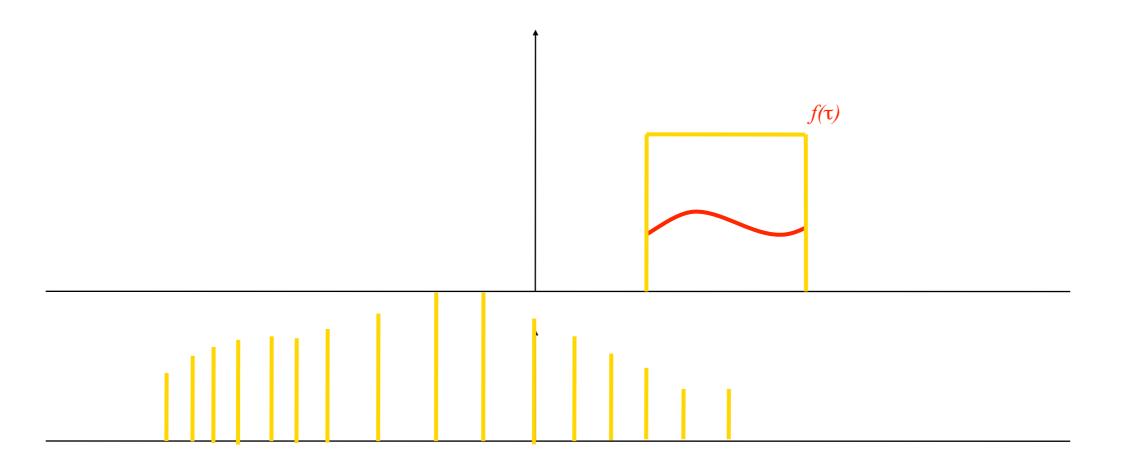






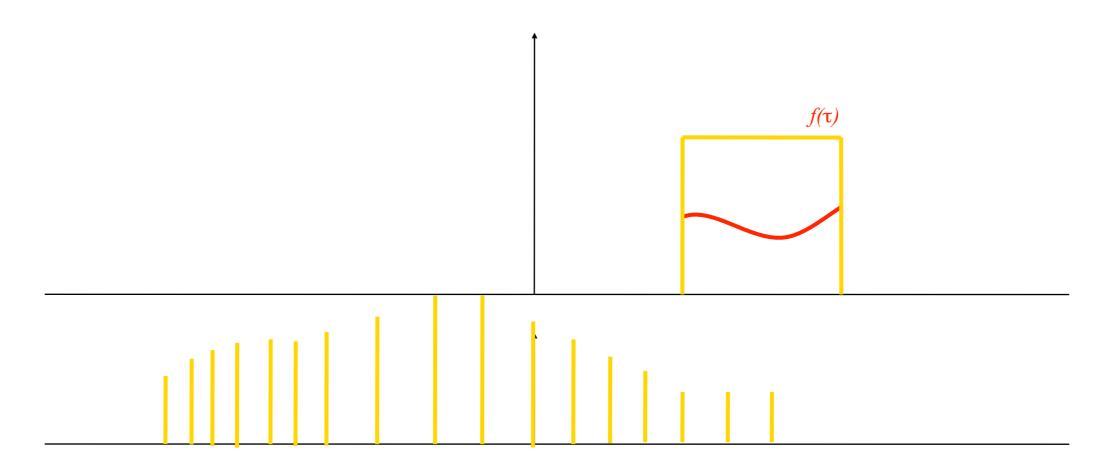






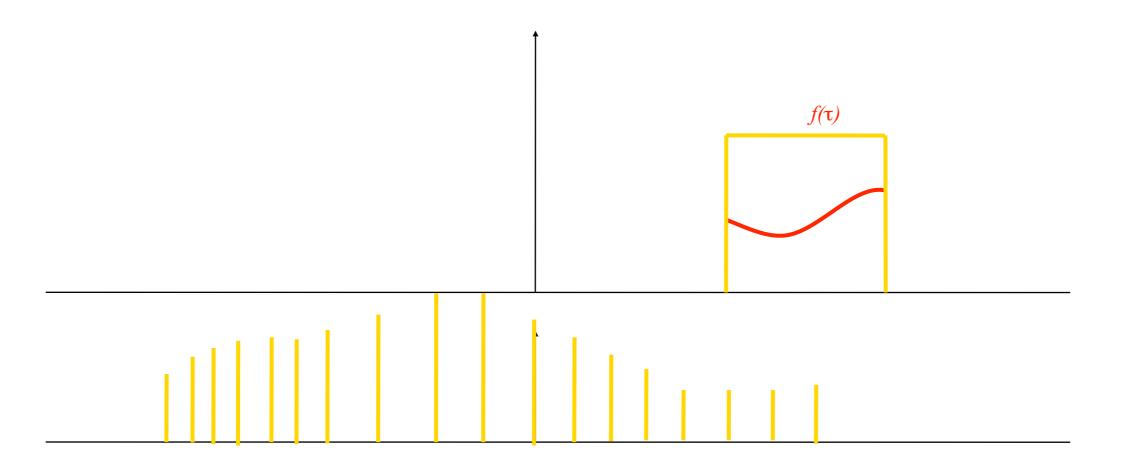






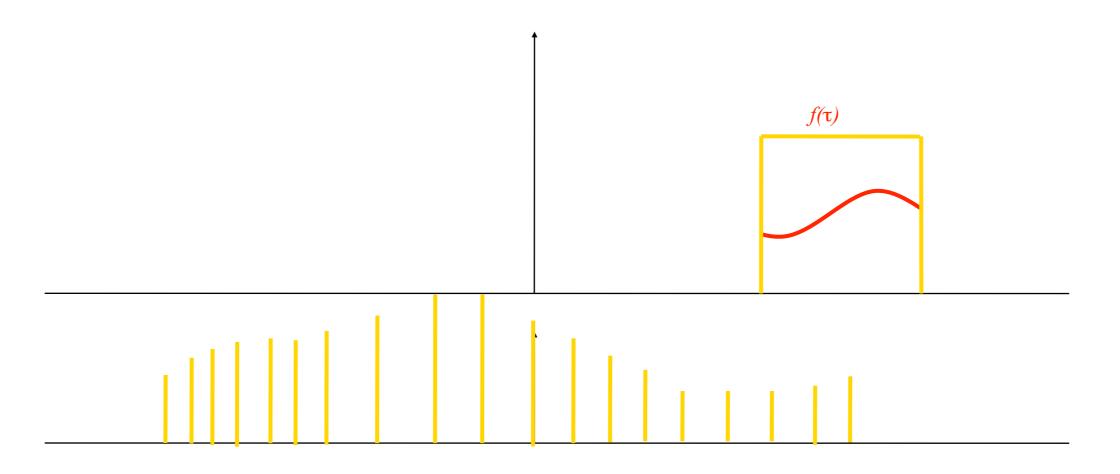






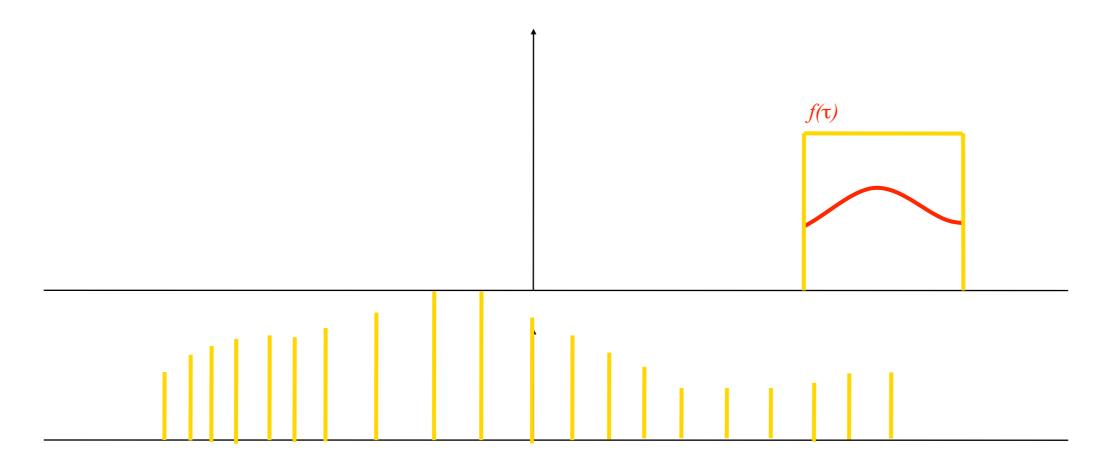






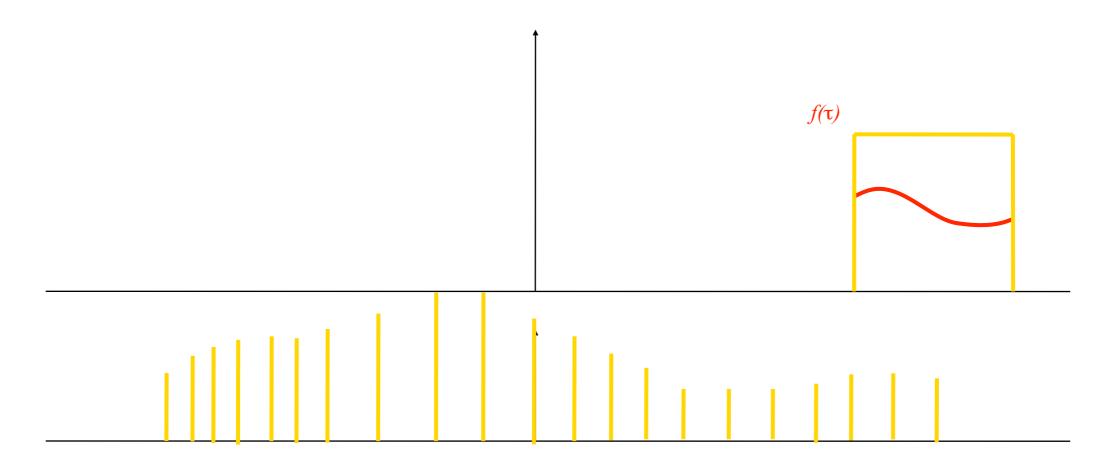






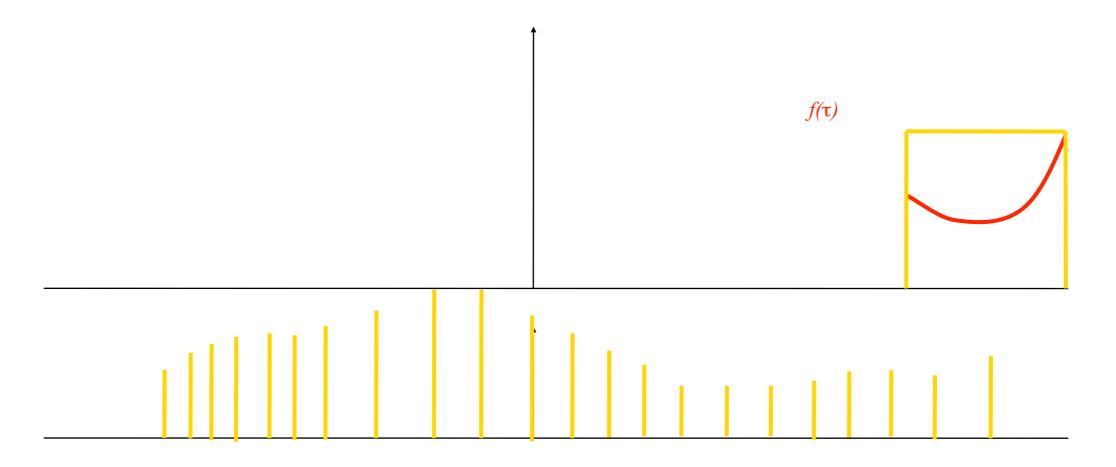








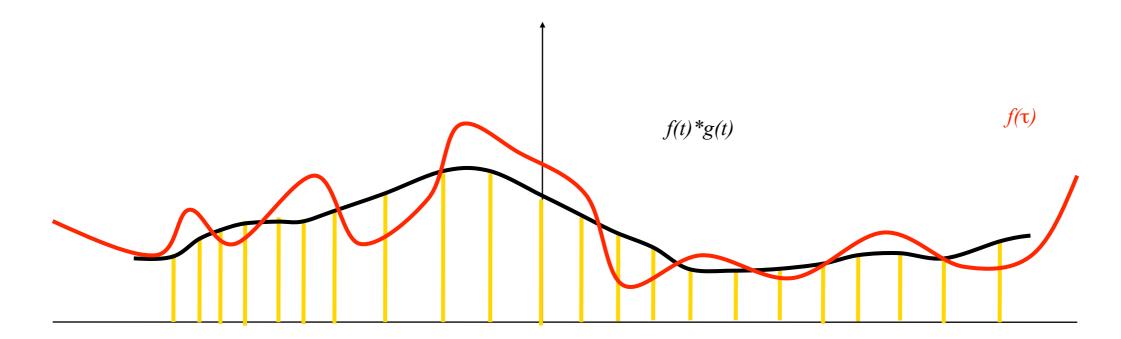








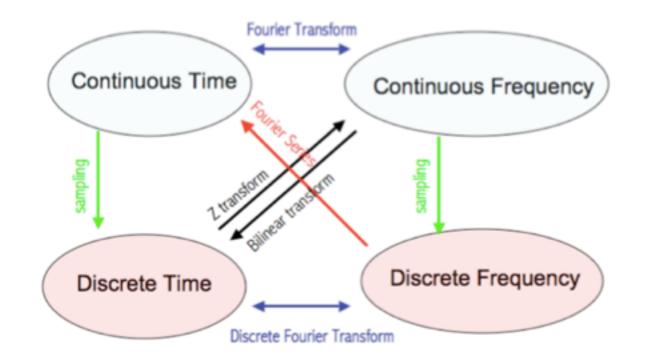
### This particular convolution smooths out some of the high frequencies in f(t).





#### Various spaces and transforms





Signal type	Continuous time	Discrete time	Transform Domain
Finite duration	Laplace	z	Continuous complex frequency (s-plane)
Finite duration	Fourier	Discrete-time Fourier (DTFT)	Continuous real frequency
Periodic	Fourier Series	Discrete Fourier Series (DFS)	Discrete real frequency





Figure 6.3-1: Definition of a linear system.

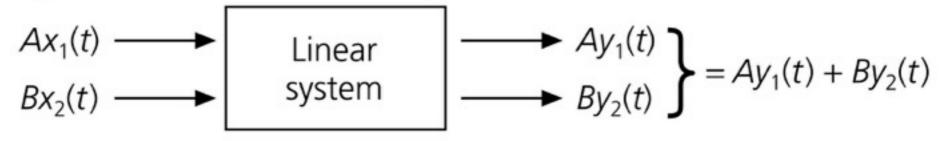
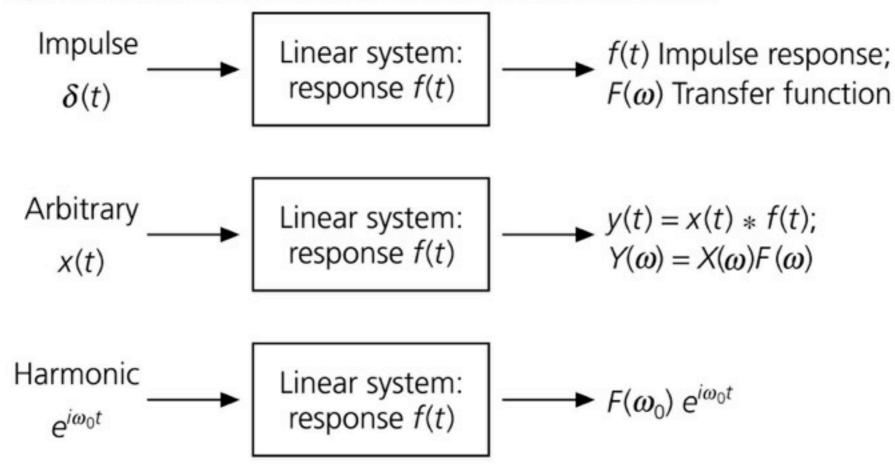
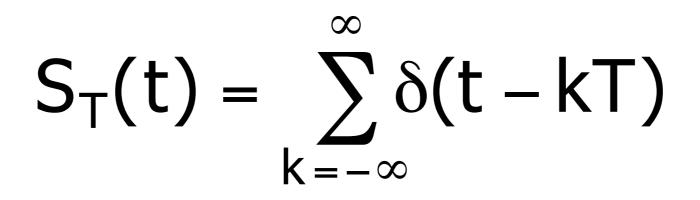


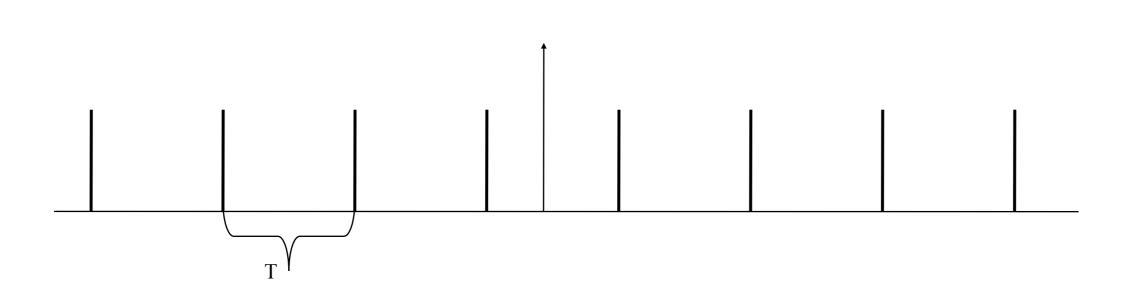
Figure 3.3-29: Seismic section before and after deconvolution.





### A Sampling Function or Impulse Train is defined by:









The Fourier Transform of the Sampling Function is itself a sampling function.

The sample spacing is the inverse.

# $S_{T}(t) \Leftrightarrow S_{\frac{1}{T}}(\omega)$





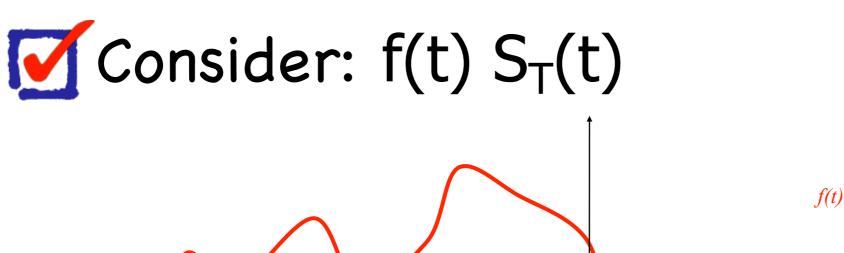
The convolution theorem states that convolution in the spatial domain is equivalent to multiplication in the frequency domain, and viceversa.

 $f(t) * g(t) \Leftrightarrow F(\omega)G(\omega)$  $f(t)g(t) \Leftrightarrow F(\omega) * G(\omega)$ 





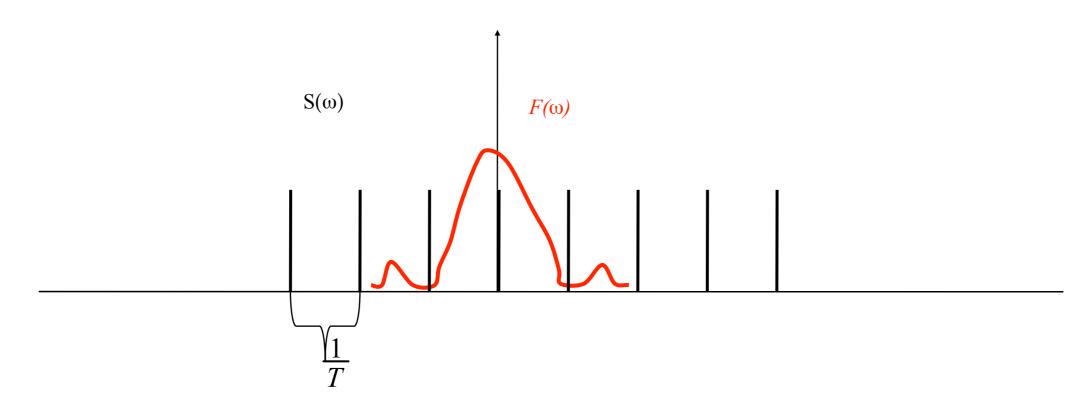
### This powerful theorem can illustrate the problems with our point sampling and provide guidance on avoiding aliasing.







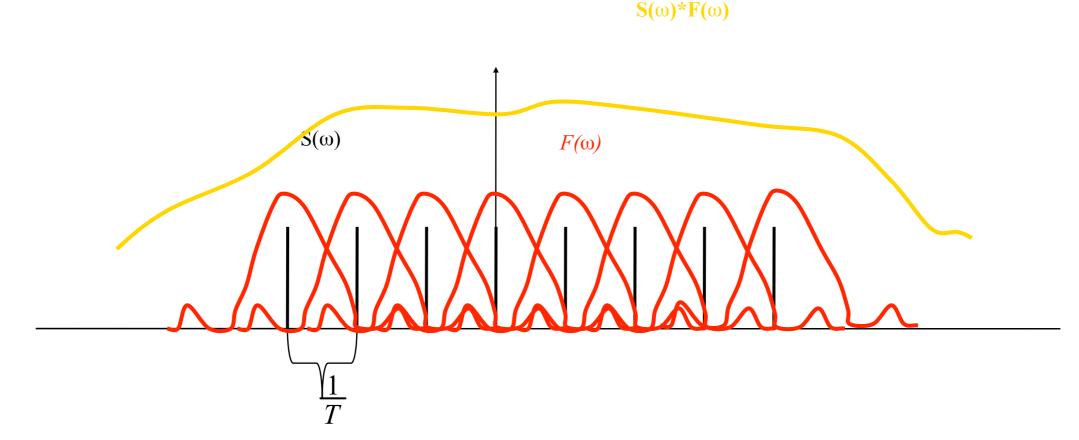
### What does this look like in the Fourier domain?







### In Fourier domain we would convolve







What this says, is that any frequencies greater than a certain amount will appear intermixed with other frequencies.

In particular, the higher frequencies for the copy at 1/T intermix with the low frequencies centered at the origin.





Note, that the sampling process introduces frequencies out to infinity.

- We have also lost the function f(t), and now have only the discrete samples.
- This brings us to our next powerful theory.





The Shannon Sampling Theorem A band-limited signal f(t), with a cutoff frequency of  $\lambda$ , that is sampled with a sampling spacing of T may be perfectly reconstructed from the discrete values f[nT] by convolution with the sinc(t) function, provided: 1

$$\lambda < \frac{1}{2T}$$





### Why is this?

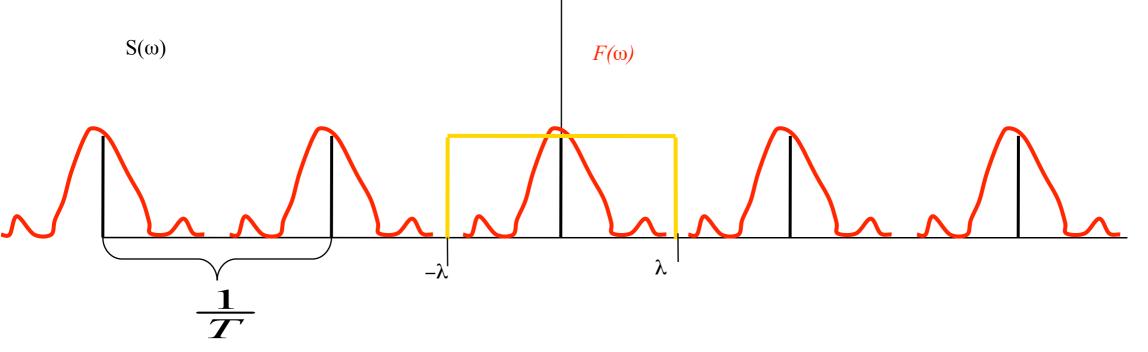
- Moreover the timit will ensure that the copies of  $F(\omega)$  do not overlap in the frequency domain.
- I can completely reconstruct or determine f(t) from F(ω) using the Inverse Fourier Transform.





## In order to do this, I need to remove all of the shifted copies of $F(\omega)$ first.

This is done by simply multiplying  $F(\omega)$  by a box function of width  $2\lambda$ .

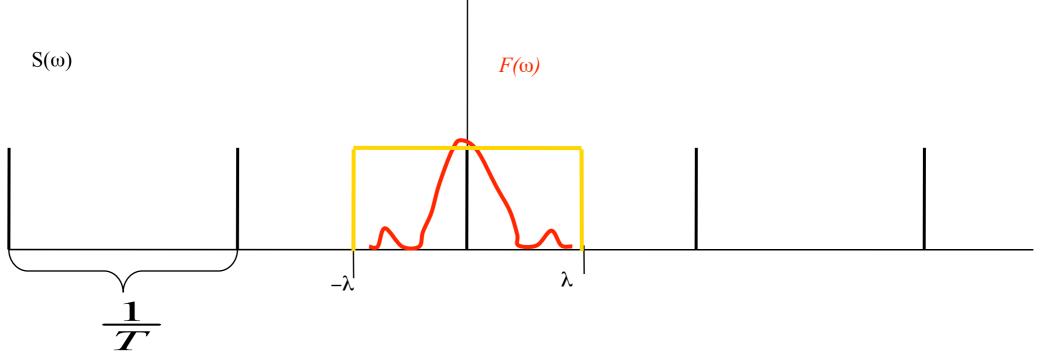






# In order to do this, I need to remove all of the shifted copies of $F(\omega)$ first.

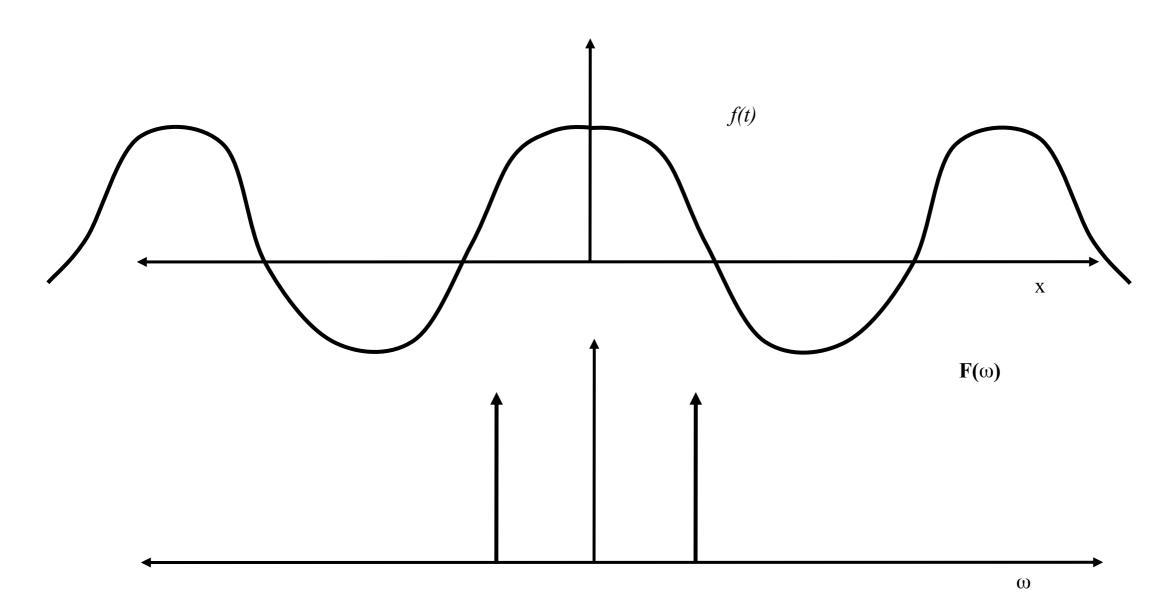
## This is done by simply multiplying $F(\omega)$ by a box function of width $2\lambda$ .







### Consider the function $f(t) = cos(2\pi t)$ .







So, given f[nT] and an assumption that f(t) does not have frequencies greater than 1/2T, we can write the formula:

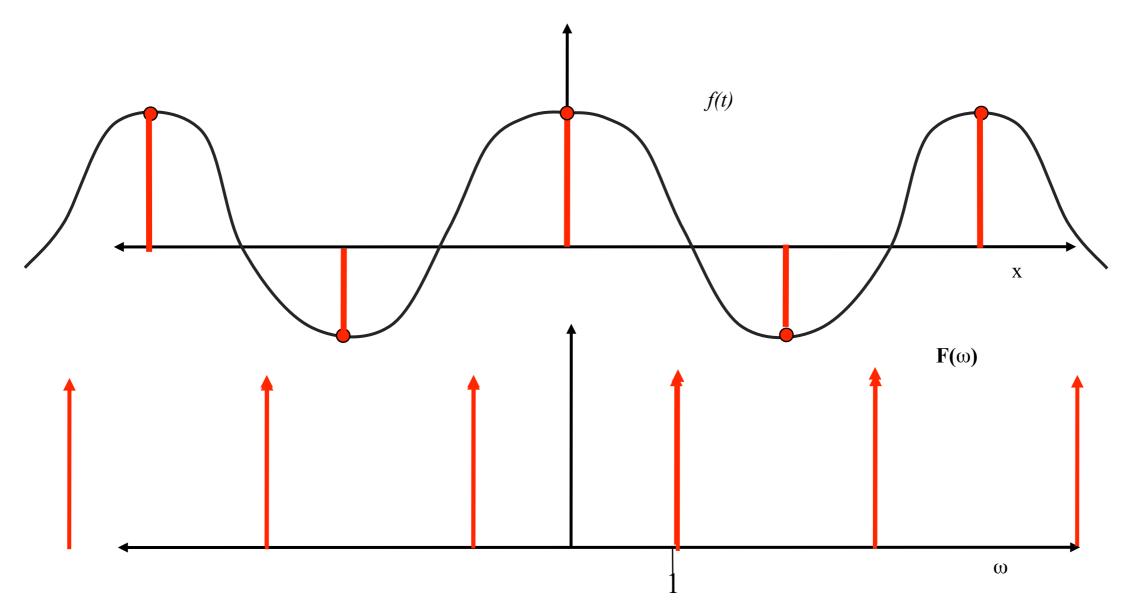
 $f[nT] = f(t) S_{T}(t) \Leftrightarrow F(\omega)^{*} S_{T}(\omega)$   $F(\omega) = (F(\omega)^{*} S_{T}(\omega)) Box_{1/2T}(\omega)$   $f(t) = f[nT]^{*} sinc(t)$ 

f(t) = f[nT] \* sinc(t)





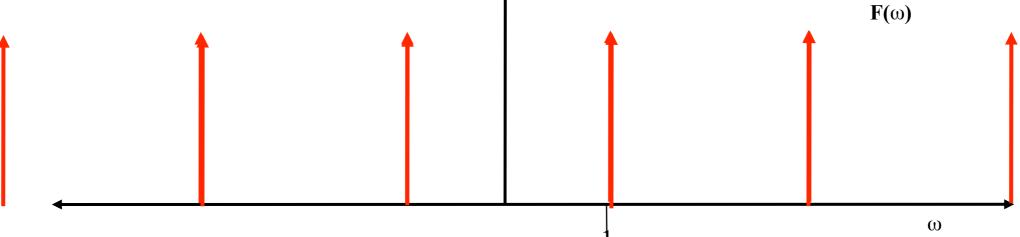
### Mow sample it at T=1/2







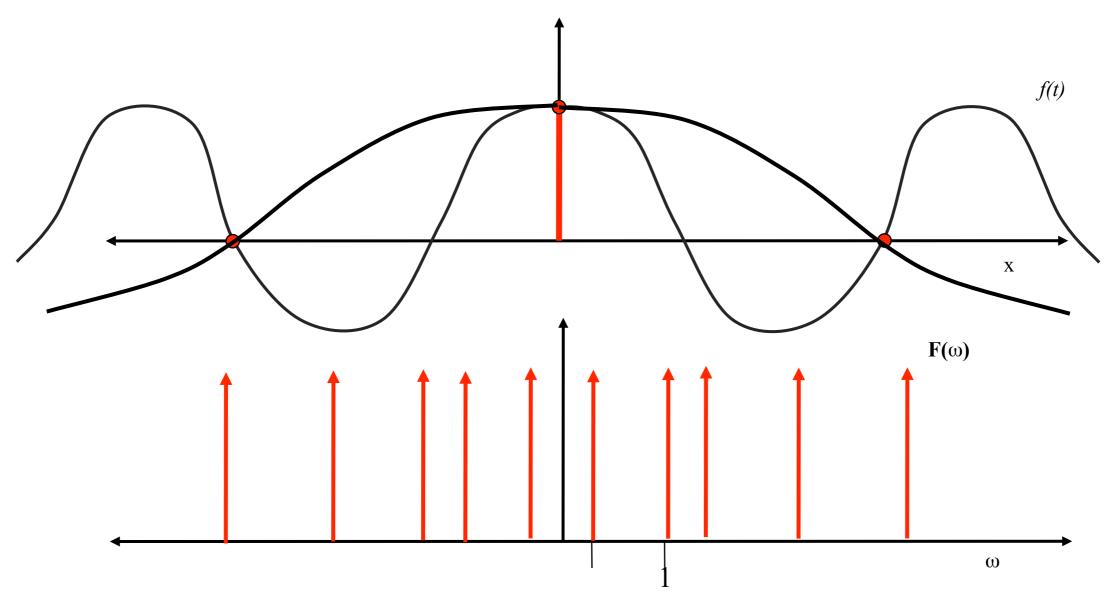
**Problem**: The amplitude is now wrong or undefined. Mote however, that there is one and only one cosine with a frequency less than or equal to 1 that goes through the sample pts. **F**(ω)







#### What if we sample at T=2/3?

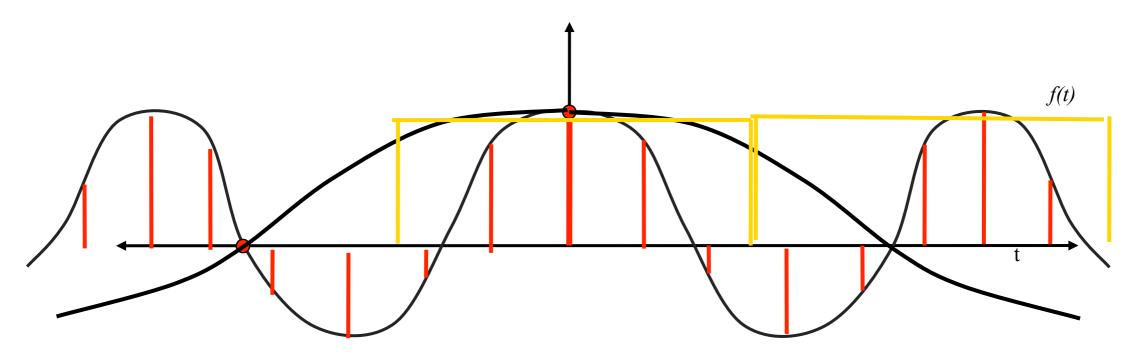


Linear Systems





Supersampling increases the sampling rate, and then integrates or convolves with a box filter, which is finally followed by the output sampling function.







### The problem:

- The signal is not band-limited.
- Uniform sampling can pick-up higher frequency patterns and represent them as low-frequency patterns.

