Signal Processing III (Discrete Fourier Transform)

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Fourier Integral Pair (Fourier Transform)

$$x(t) = \int_{-\infty}^{\infty} X(f) e^{i2\pi f t} df \qquad X(f) = \int_{-\infty}^{\infty} x(t) e^{-i2\pi f t} dt$$

The Fourier transform (FT) decomposes a function of continuous time (a signal) into its constituent frequencies. Inverse Fourier transform is to reconstruct the signals using frequencies and its amplitude.

Properties of Fourier Transforms

1. Time scaling

$$F\big(x(at)\big) = \frac{1}{|a|} X\left(\frac{f}{a}\right)$$

Q: What does this mean by?

2. Time reversal

$$F\big(x(-t)\big) = X(-f)$$

3. Differentiation

$$F\big(\dot{x}(t)\big) = i2\pi f X(f)$$

4. Time shifting

$$F\big(x(t-t_0)\big) = e^{-i2\pi f t_0} X(f)$$

Only phase shift !

5. Modulation

$$F\big(x(t)e^{i2\pi f_0 t}\big) = X(f-f_0)$$

$$F(x(t)\cos(2\pi f_0 t)\,)$$
$$= \frac{1}{2}[X(f-f_0) + X(f+f_0)\,]$$

Sine wave

Properties of Fourier Transforms (Convolution)

$$F(h(t) * x(t)) = H(f)X(f)$$

$$F(x(t)w(t)) = X(f) * W(f)$$

Where the convolution of the two functions $h(t)$ and $x(t)$ is defined as

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

It is a function of time !

Proof of Convolution Relationship

Supplement

$$F(h(t) * x(t)) = H(f)X(f)$$

Proof:

$$F\big(h(t) * x(t)\big) = \int_{-\infty}^{\infty}\left[\int_{-\infty}^{\infty} h(\tau)x(t-\tau)\,d\tau\right]\exp^{-i2\pi ft}dt$$

Fubini's Theorem

$$= \int_{-\infty}^{\infty}\int_{-\infty}^{\infty} h(\tau)x(v)\exp^{-i2\pi f(\tau+v)}\,d\tau\,dv \qquad v = t-\tau$$

$$= \int_{-\infty}^{\infty} h(\tau)\exp^{-i2\pi f(\tau)}\,d\tau\int_{-\infty}^{\infty} x(v)\exp^{-i2\pi f(v)}\,dv = H(f)X(f)$$

$$F(x(t)) = X(f) = \int_{-\infty}^{\infty} x(t)e^{-i2\pi ft}dt$$

Fourier integral

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

Conovlution

Proof of Convolution Relationship (Continue)

Supplement

$$F(x(t)w(t)) = X(f) * W(f)$$

Proof: $F(x(t)w(t)) = \int_{-\infty}^{\infty} x(t)\, w(t) \exp^{-i2\pi ft}\, dt$

$$= \int_{-\infty}^{\infty}\int_{-\infty}^{\infty}\int_{-\infty}^{\infty} X(f_1) \exp^{i2\pi f_1 t}\, W(f_2) \exp^{i2\pi f_2 t} \exp^{-i2\pi ft}\, df_1\, df_2\, dt$$

$$= \int_{-\infty}^{\infty} X(f_1) \int_{-\infty}^{\infty} W(f_2) \int_{-\infty}^{\infty} \exp^{-i2\pi(f-f_1-f_2)t}\, dt\, df_2\, df_1$$

$$= \int_{-\infty}^{\infty} X(f_1) \int_{-\infty}^{\infty} W(f_2)\delta(f-f_1-f_2) df_2 df_1 = \int_{-\infty}^{\infty} X(f_1) W(f-f_1) df_1 = X(f) * W(f)$$

$$F(x(t)) = X(f) = \int_{-\infty}^{\infty} x(t) e^{-i2\pi ft} dt$$

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

$$x(t) = \int_{-\infty}^{\infty} X(f) e^{i2\pi ft} df$$

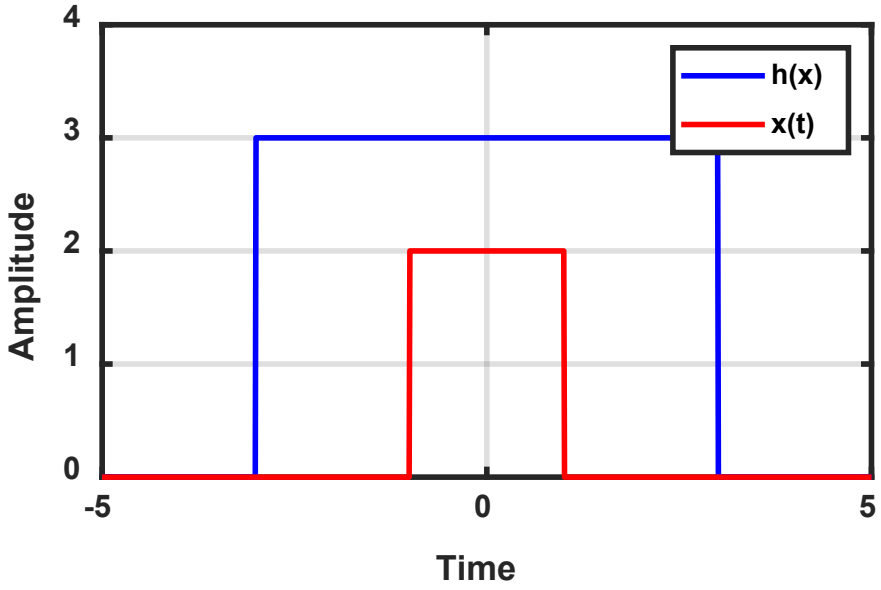
Example: Convolution 1 (Square Wave)

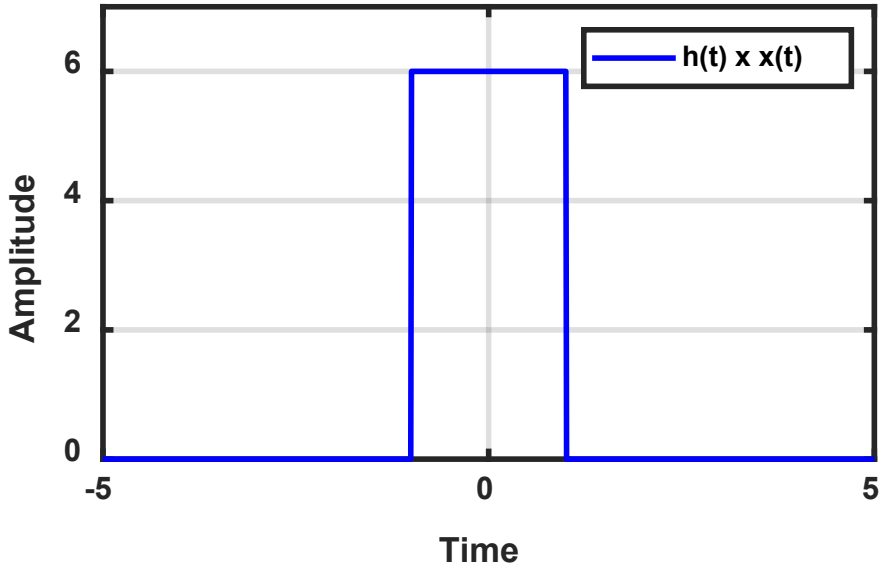
$h(t) = 3 \quad if \ -3 < t < 3$. Otherwise, $h(t) = 0$

$x(t) = 2 \quad if \ -1 < t < 1$. Otherwise, $x(t) = 0$

$$h(t)x(t)$$

Multiply values of *h* and *x* at each time point

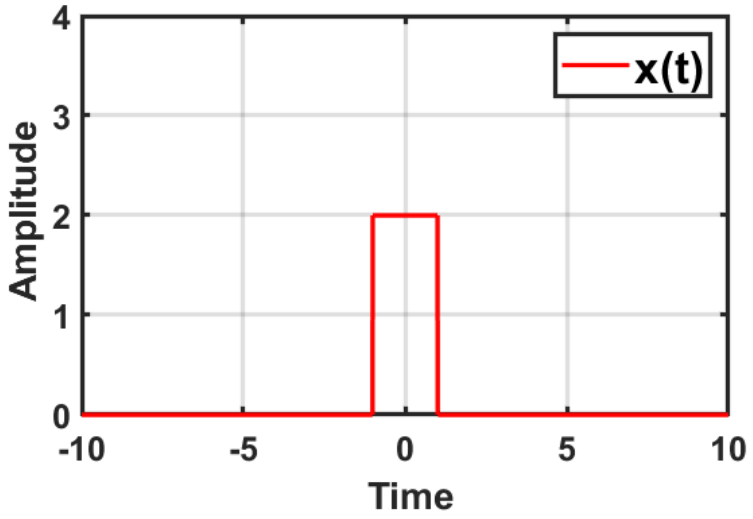


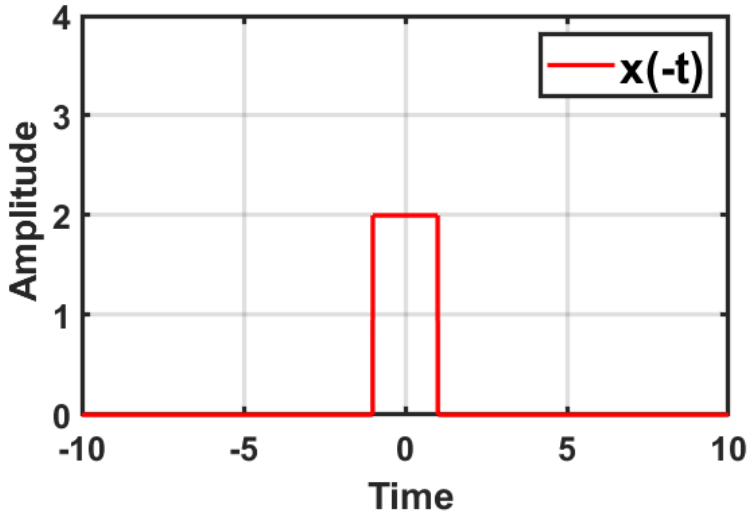


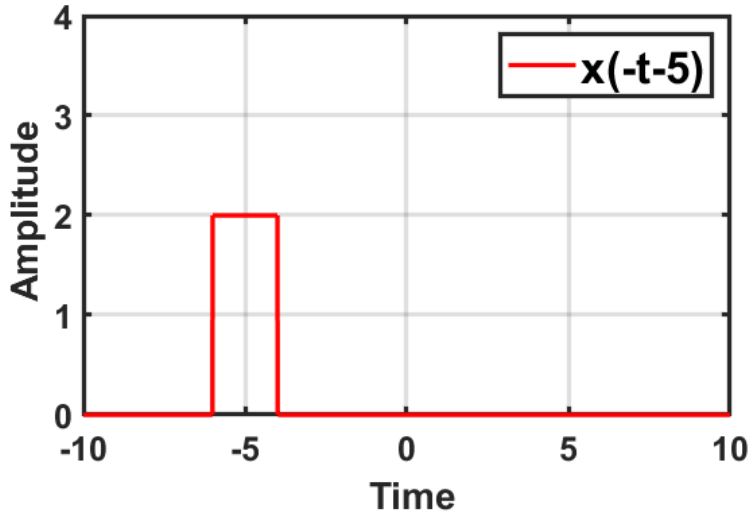
Tutorial

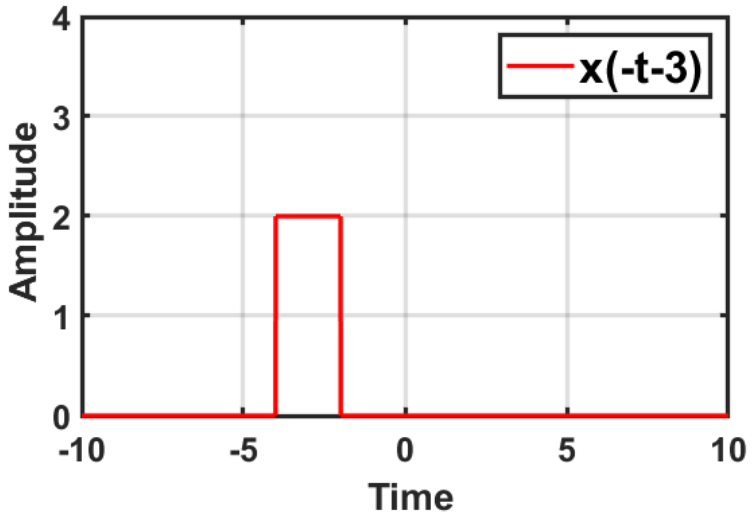
Time Shifting

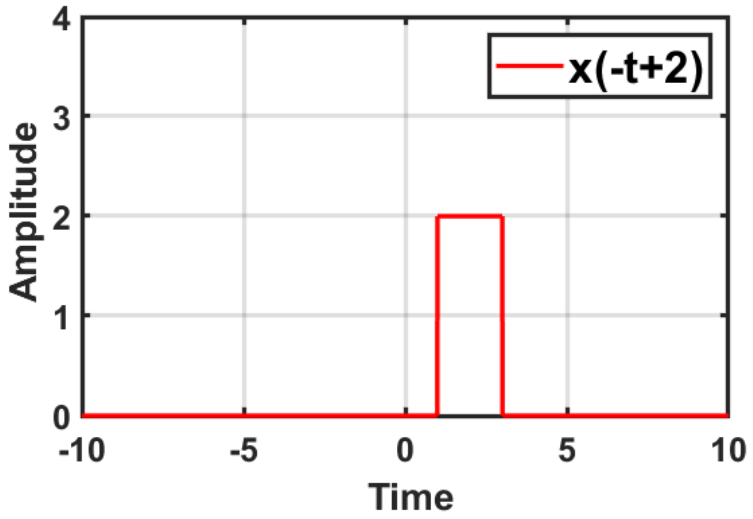
Supplement

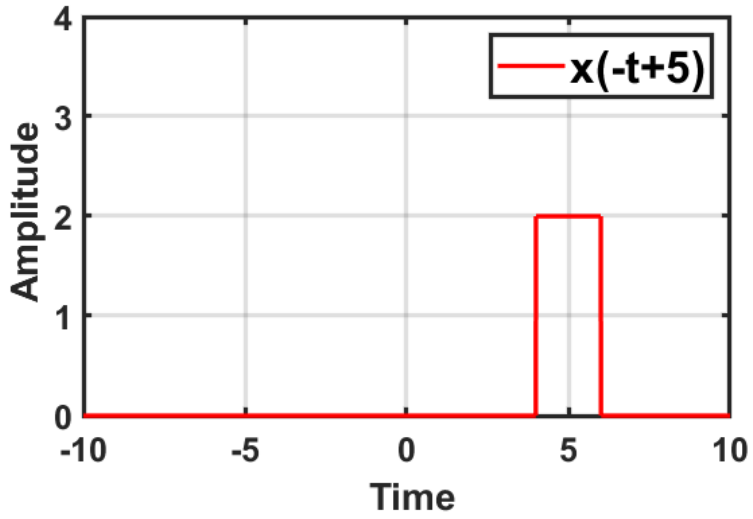












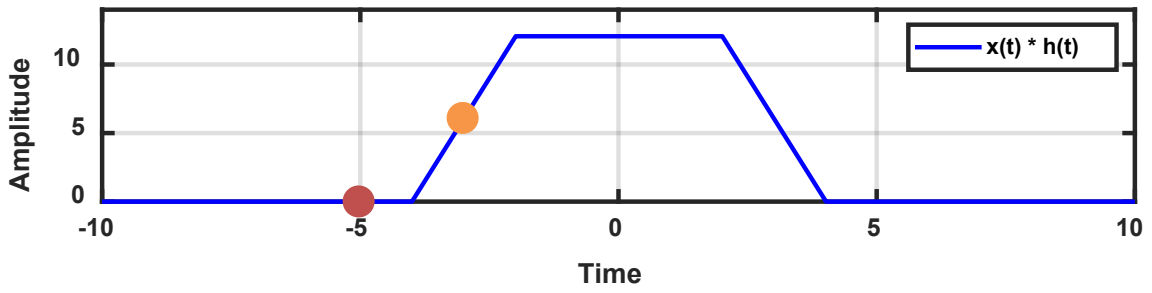
Tutorial

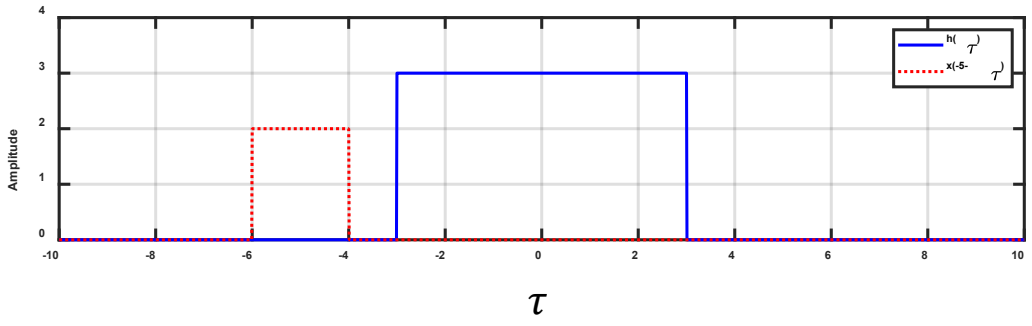
Example: Convolution 1 (Square Wave) (Continue)

Tutorial

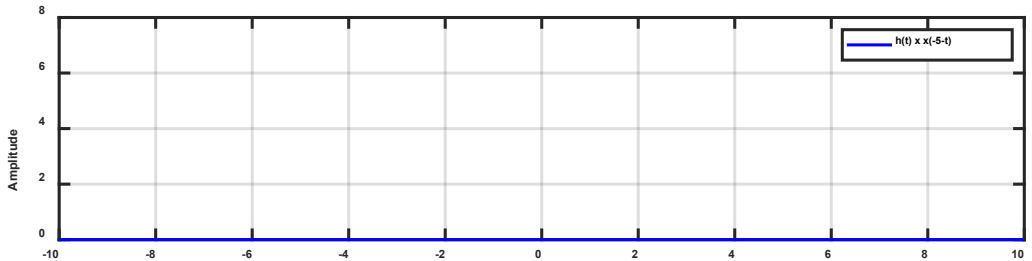
$$h(t) * x(t) = \int_{-\infty}^{\infty} h(\tau)x(t-\tau)d\tau$$

Convolution





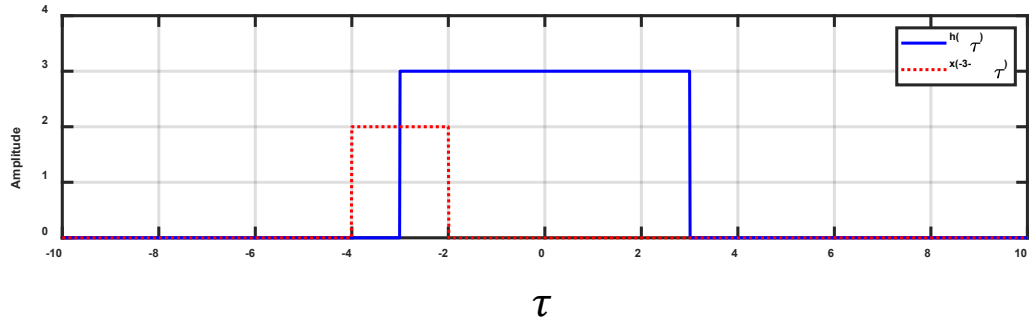
τ



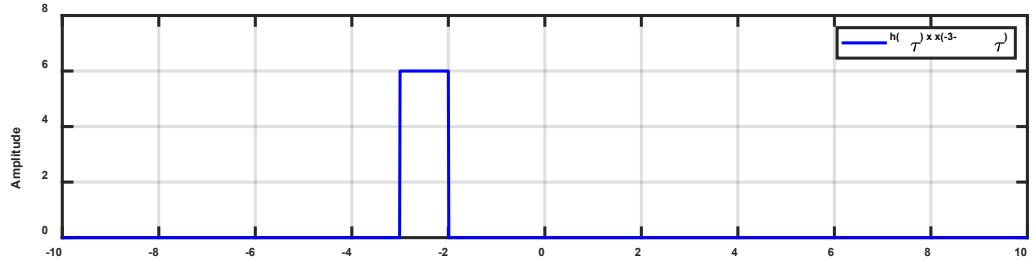
τ

$$\int_{-\infty}^{\infty} h(\tau)x(-5-\tau)\,d\tau = 0$$

It is for finding a value of x(t)*h(t) at t= -5



τ



τ

$$\int_{-\infty}^{\infty} h(\tau)x(-3-\tau)\,d\tau = 6$$

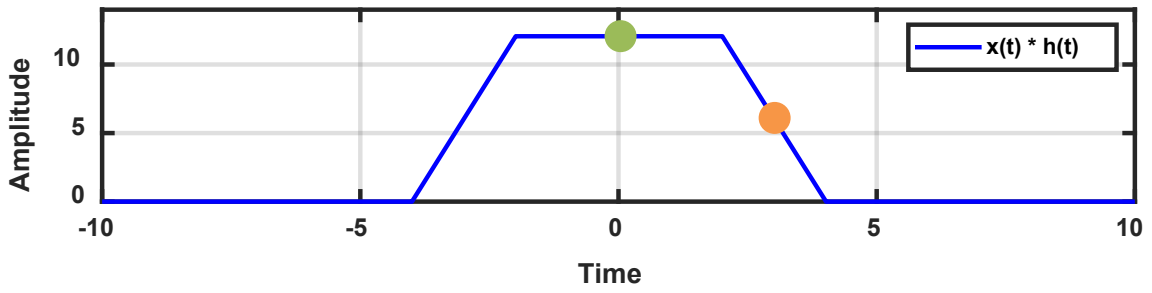
It is for finding a value of x(t)*h(t) at t= -3

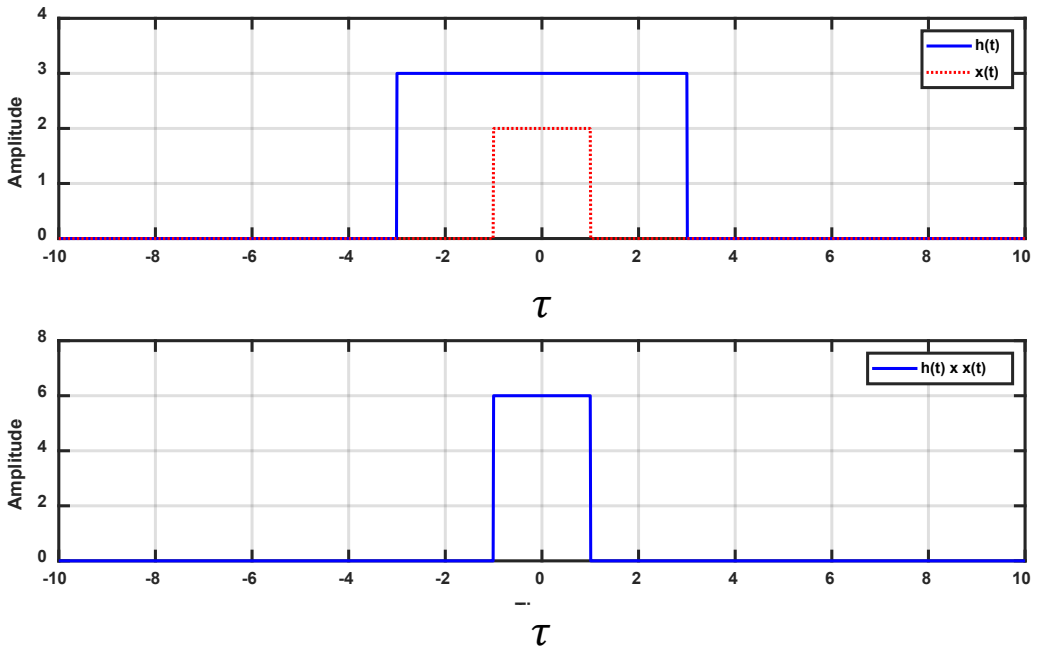
Example: Convolution 1 (Square Wave) (Continue)

Tutorial

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

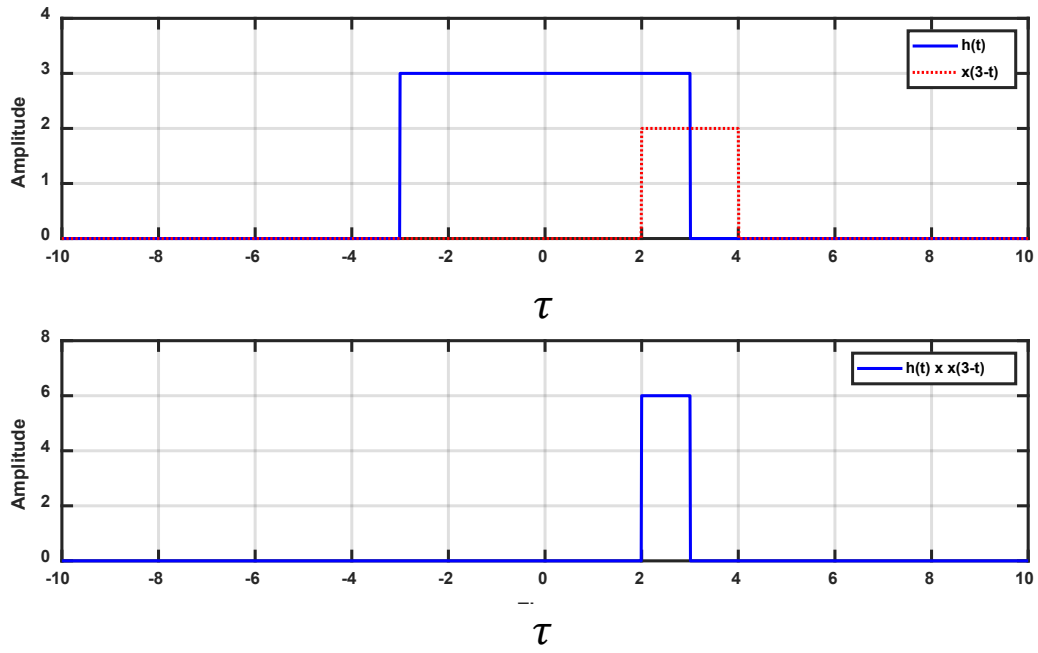
Convolution





$$\int_{-\infty}^{\infty} h(\tau)x(\tau)\,d\tau = 12$$

It is for finding a value of x(t)*h(t) at t= 0



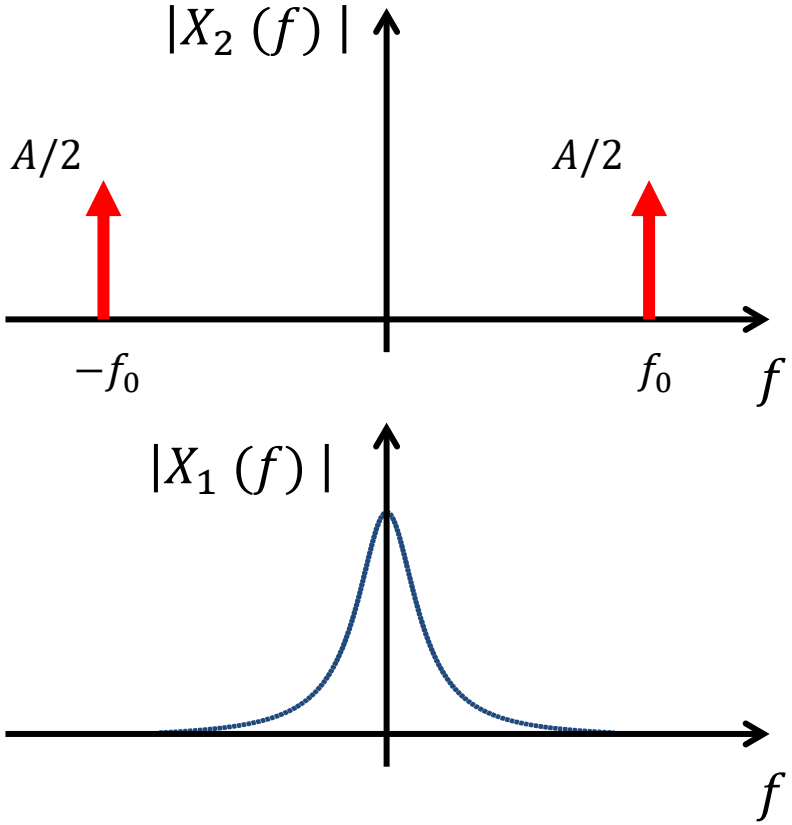
$$\int_{-\infty}^{\infty} h(\tau)x(3-\tau)\,d\tau = 6$$

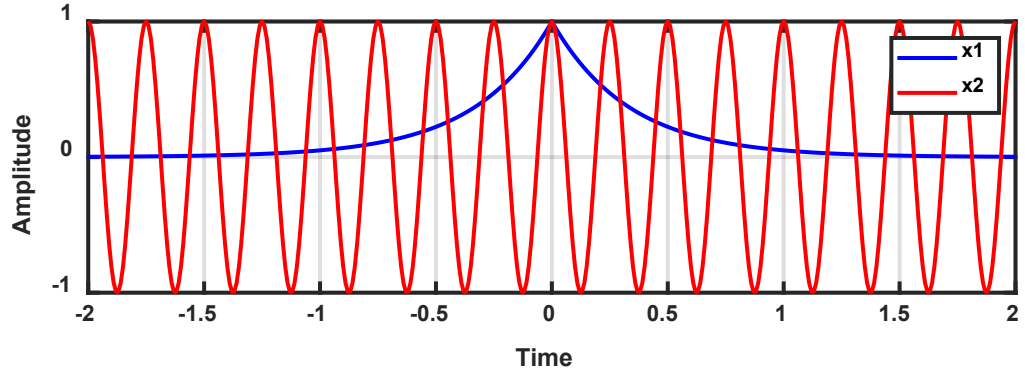
It is for finding a value of x(t)*h(t) at t= 3

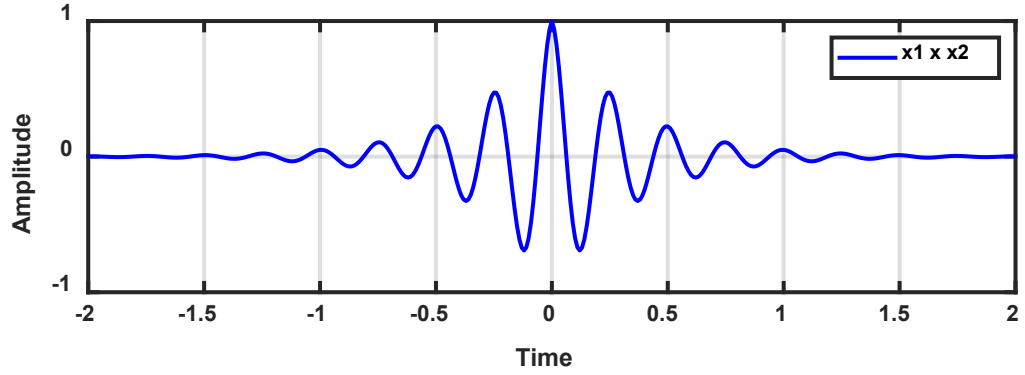
Example: Convolution 2 (Damped Symmetrically Oscillating Function)

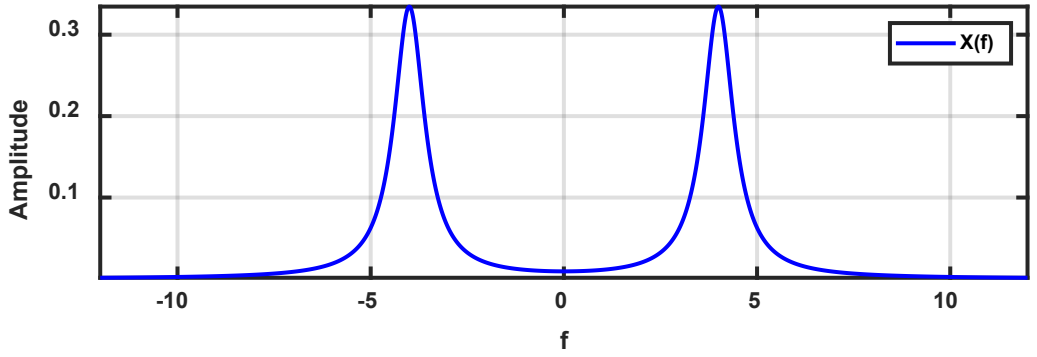
$$x_1(t) = e^{-a|t|},\ x_2(t) = cos2\pi f_0 t$$

$$x(t) = x_1(t)x_2(t) = e^{-a|t|}cos2\pi f_0 t$$

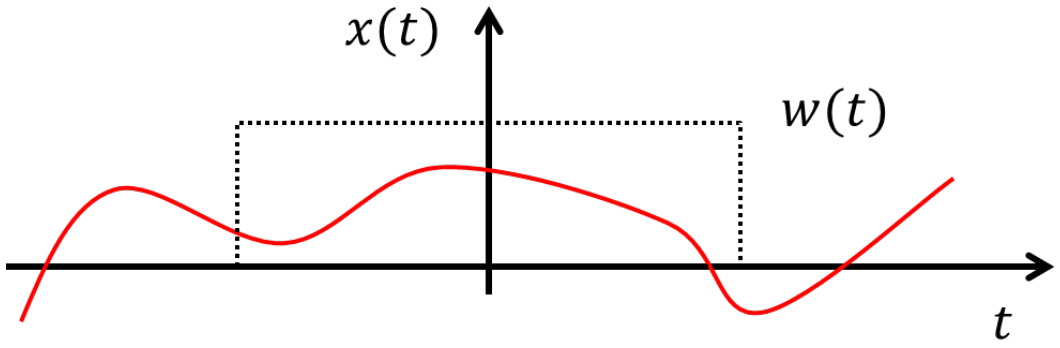








Effect of Data Truncation (Windowing) – Finite Signal



$w(t) = 1$ for $|t| <$T/2
$= 0$ otherwise

Q: How to represent a finite signal?

$x(t)$ is known (or recorded) only for $-\frac{T}{2} < t < \frac{T}{2}$, denoted as $x_T(t)$

$$F(x_T(t)) = F(x(t)w(t)) = X(f) * W(f)$$

Fourier transform of the product of two-time signals is the convolution of their Fourier transforms.

Revisit: Continuous Signal

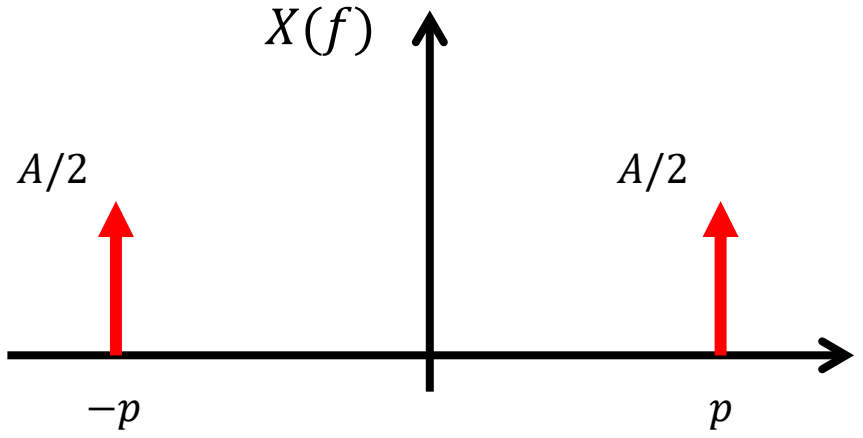
Supplement

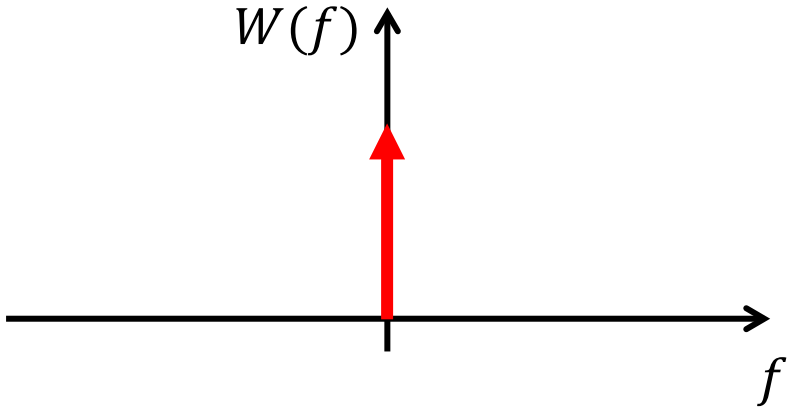
$$x(t) = A\cos 2\pi pt$$

$$X(f) = \frac{A}{2}[\delta(f-p) - \delta(f+p)]$$

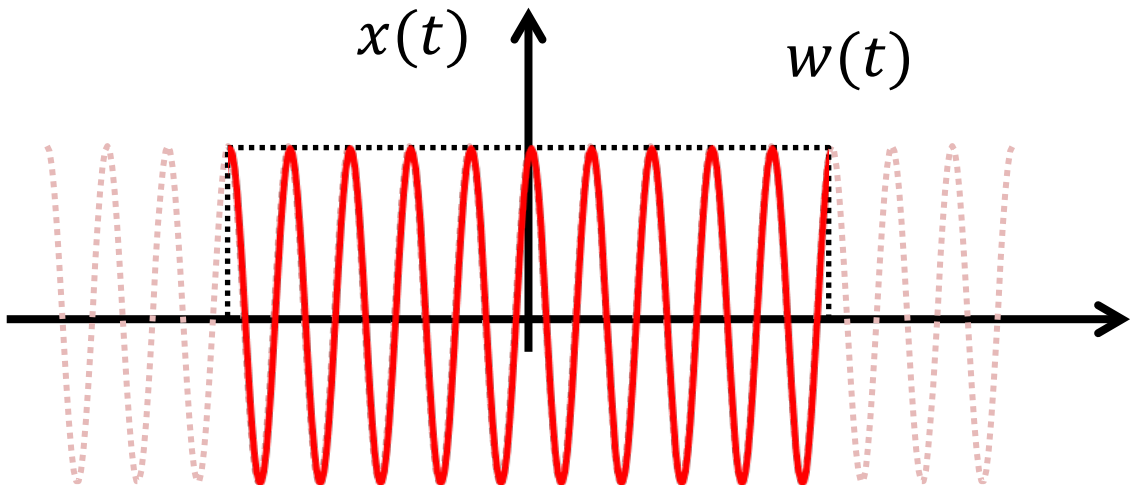
$$w(t) = 1$$

$$W(f) = \delta(f)$$

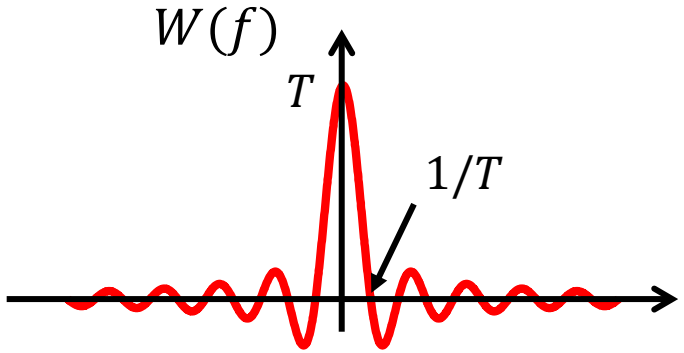




Truncated Sine Wave

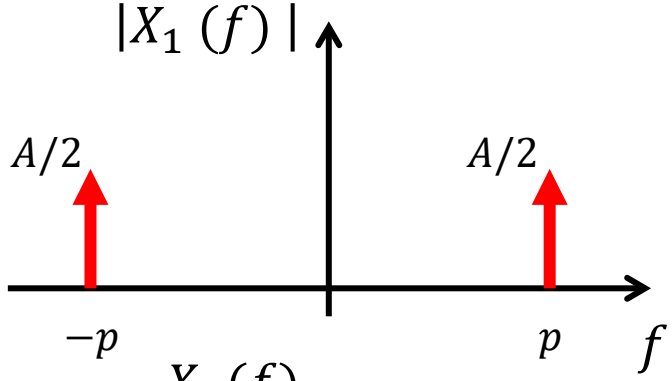


$$W(f) = \frac{T \sin \pi f T}{\pi f T}$$



$$x(t) = A \cos 2\pi p t$$

$$X(f) = \frac{A}{2}[\delta(f-p) - \delta(f+p)]$$

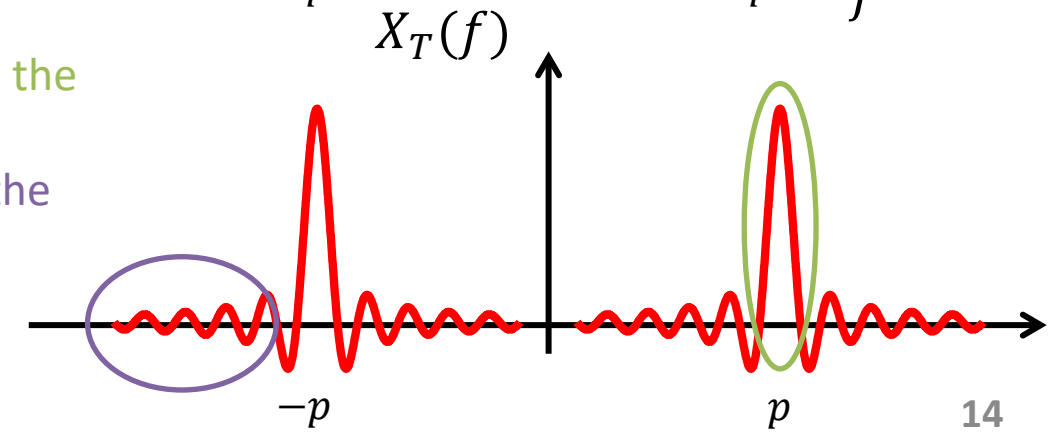


$$w(t) = 1 \quad \text{for} \quad |t| < \text{b}$$
$$= 0 \quad \text{for} \quad |t| > \text{b}$$

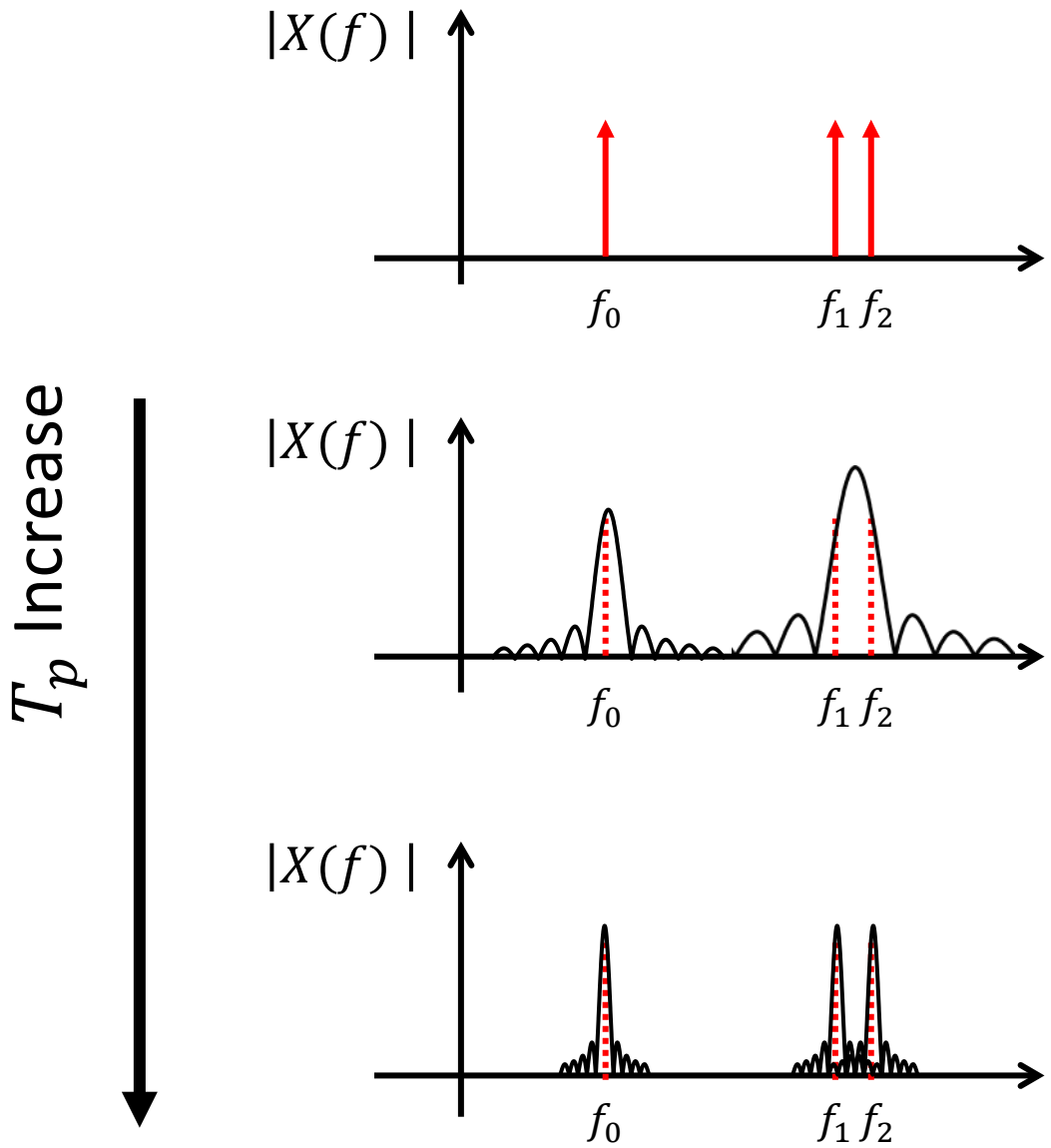
Smearing: Distortion due to the main lobe

Leakage: Distortion due to the side lobe

$$x_T(t) = x(t)w(t)$$



Effect of Data Truncation: Frequency Separation



$x(t)$ is the sum of three sine (or cosine) waves

Considerable smearing due to the spectral window

Three components are resolved but with considerable leakage at other frequencies

Window Function

Supplement

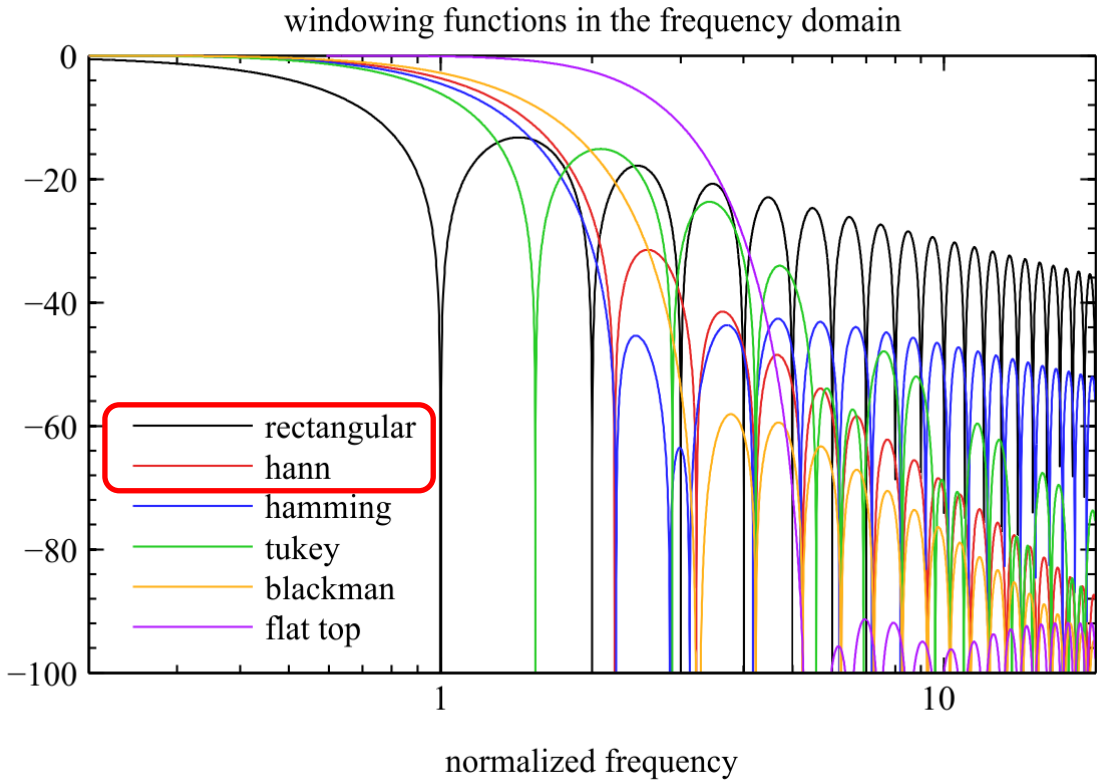
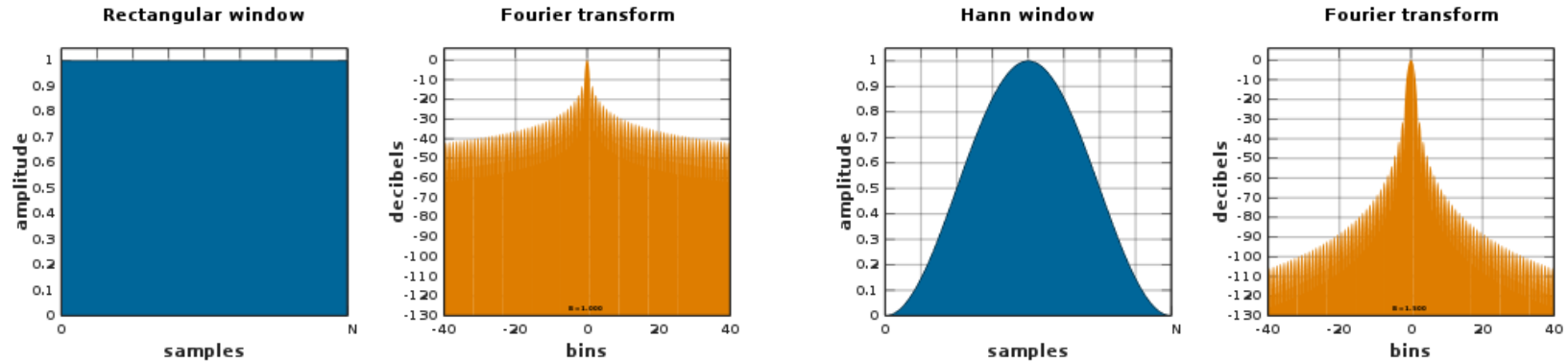
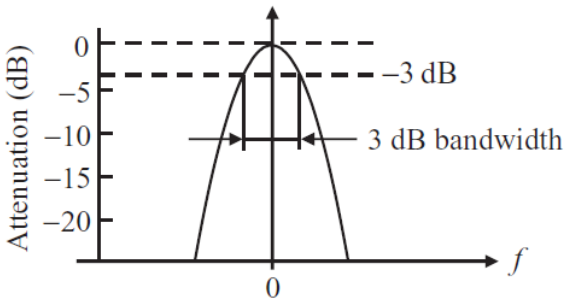


Table 4.2 Properties of some window functions

Window (length T)	Highest side lobe (dB)	Asymptotic roll-off (dB/octave)	3 dB bandwidth	Noise bandwidth	First zero crossing (freq.)
Rectangular	−13.3	6	$0.89\frac{1}{T}$	$1.00\frac{1}{T}$	$\frac{1}{T}$
Bartlett (triangle)	−26.5	12	$1.28\frac{1}{T}$	$1.33\frac{1}{T}$	$\frac{2}{T}$
Hann(ing) (Tukey or cosine squared)	−31.5	18	$1.44\frac{1}{T}$	$1.50\frac{1}{T}$	$\frac{2}{T}$
Hamming	−43	6	$1.30\frac{1}{T}$	$1.36\frac{1}{T}$	$\frac{2}{T}$
Parzen	−53	24	$1.82\frac{1}{T}$	$1.92\frac{1}{T}$	$\frac{4}{T}$

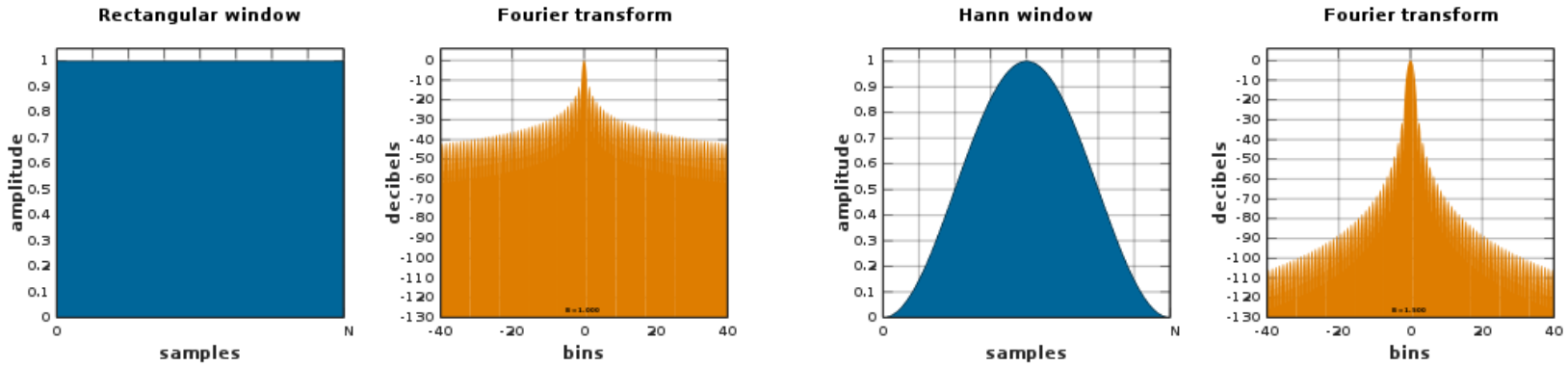




Example: Hanning Window vs Rectangular Window

$$x(t) = A_1 \sin 2\pi f_1 t + A_2 sin 2\pi f_2 t + A_3 sin 2\pi f_3 t$$

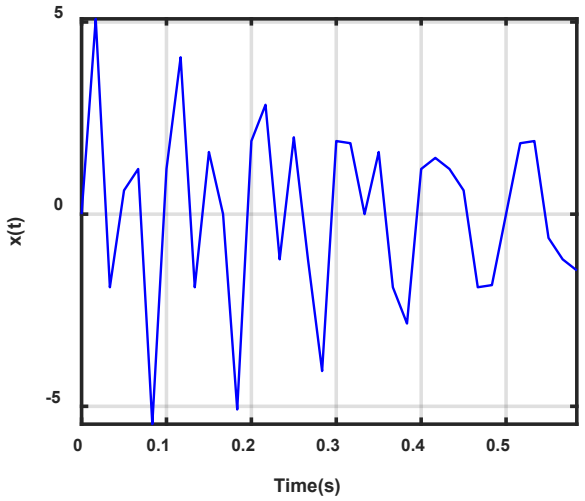
Amplitudes are $A_1 = A_2 = A_3 = 2$, which gives the magnitude '1' for each sinusoidal component in the frequency domain. The frequencies are chosen as $f_1 = 10$, $f_2 = 20$ and $f_3 = 21$.

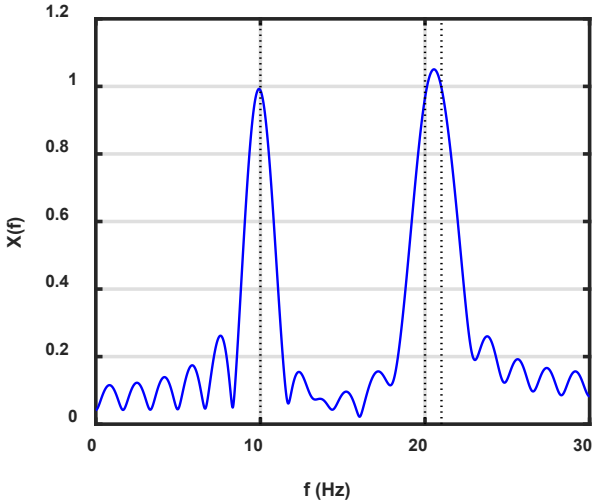


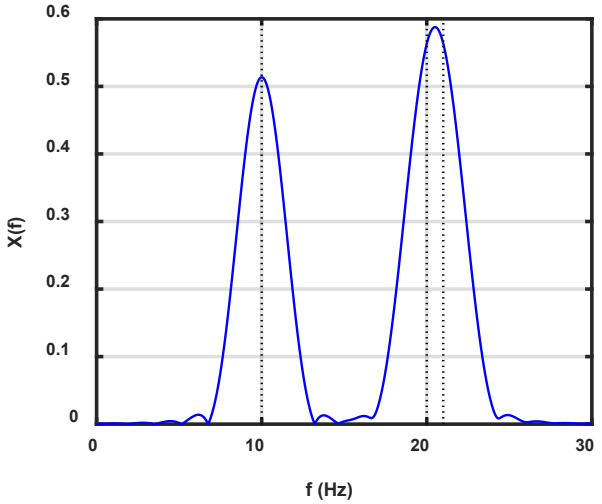
Example: Hanning Window vs Rectangular Window (Continue)

Original signal | **Rectangular** | **Hanning**

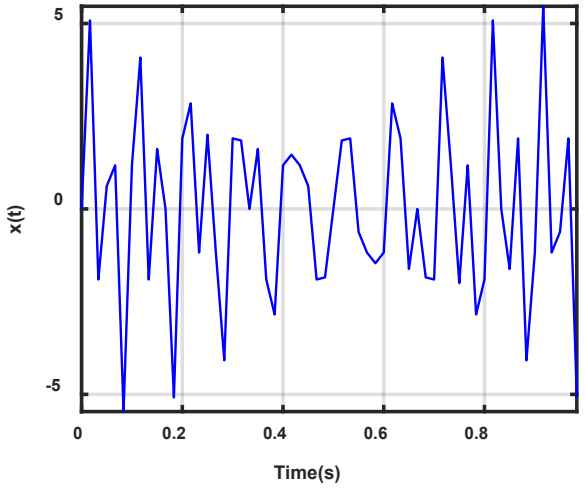
$T = 0.6s$

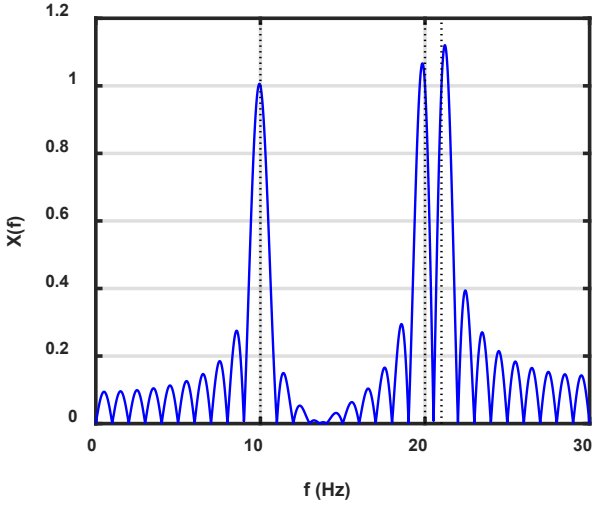


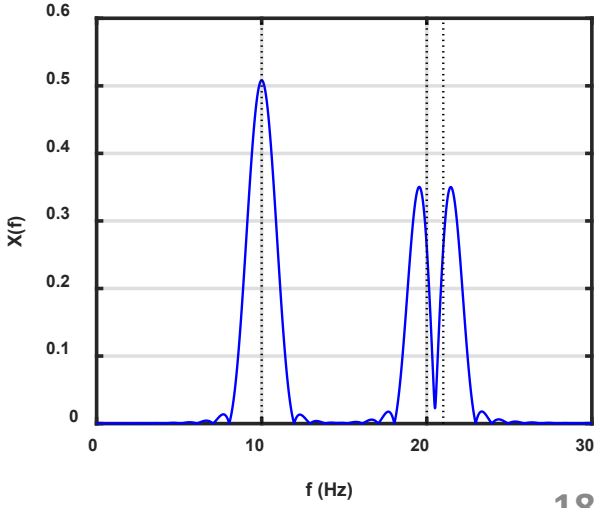




$T = 1s$





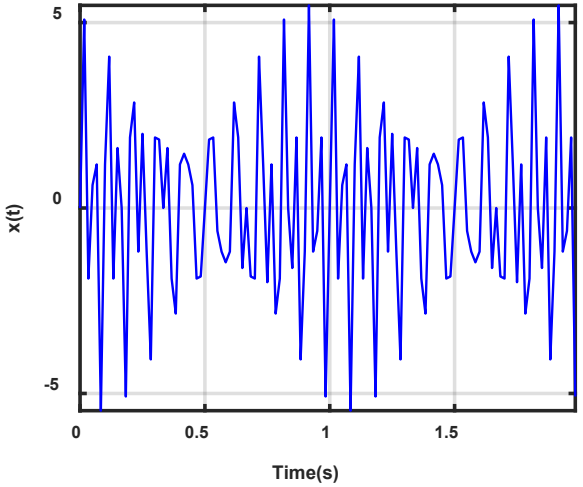


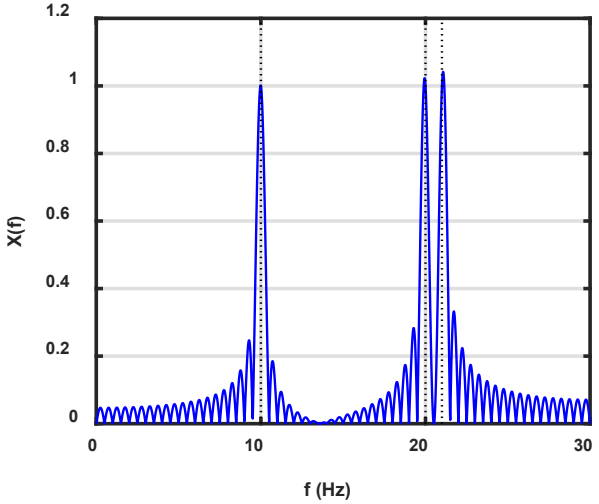
Black dotted lines indicate 10, 20, and 21 Hz

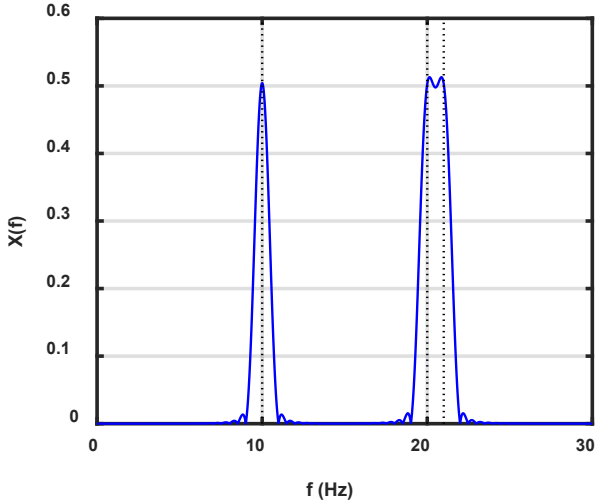
Example: Hanning Window vs Rectangular Window (Continue)

Original signal | **Rectangular** | **Hanning**

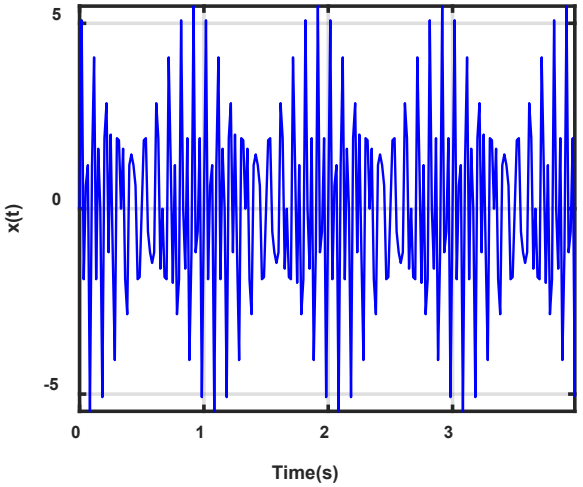
$T = 2s$

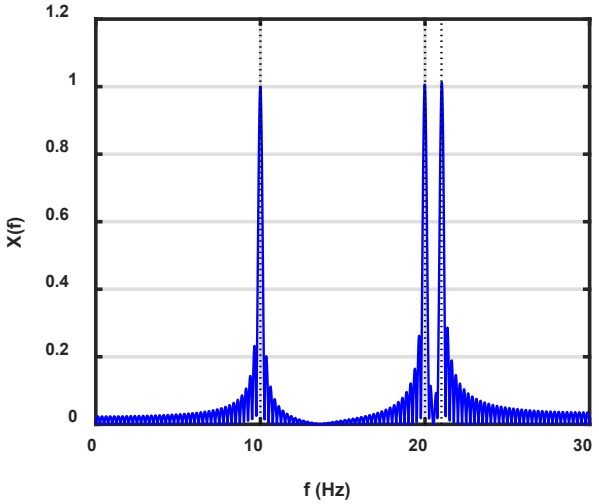


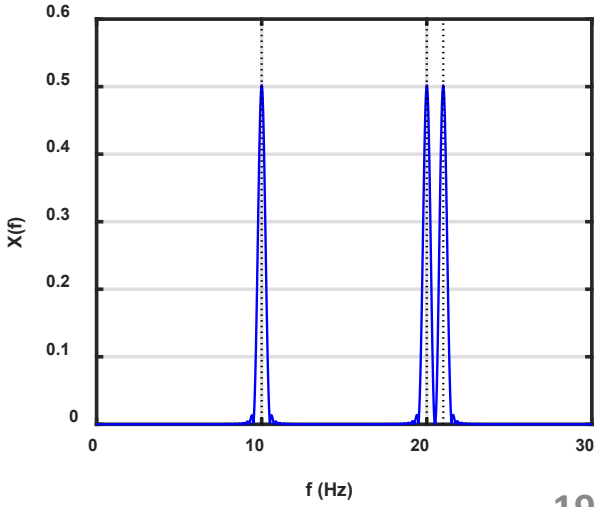




$T = 4s$







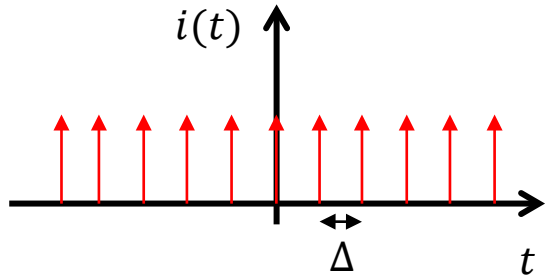
Black dotted lines indicate 10, 20, and 21 Hz

Tutorial

Impluse Train Modulation (Discretization)

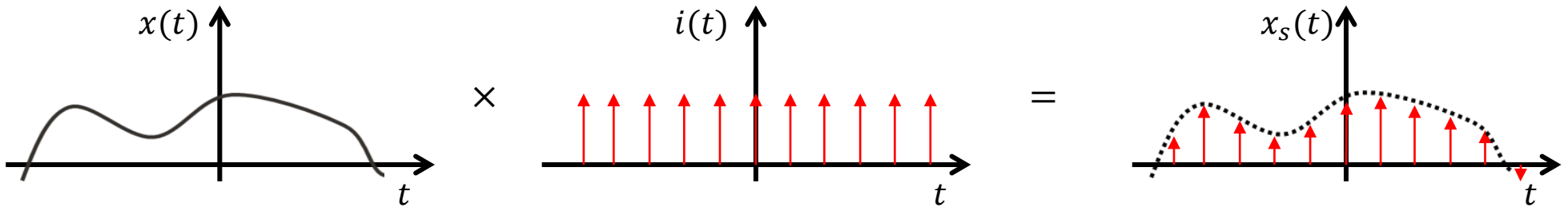
A 'train' of delta functions $i(t)$ is expressed as

$$i(t) = \sum_{n=-\infty}^{\infty} \delta(t - n\Delta)$$



$$f_S = \frac{1}{\Delta}$$

The sampling procedure can be illustrated as $x_s(t) = x(t)i(t) = x(n\Delta)$



$$X_s(f) = \int_{-\infty}^{\infty} \left[x(t) \sum_{n=-\infty}^{\infty} \delta(t - n\Delta) \right] e^{-i2\pi ft} dt = \sum_{n=-\infty}^{\infty} \left[\int_{-\infty}^{\infty} x(t) e^{-i2\pi ft} \delta(t - n\Delta) dt \right] = \sum_{n=-\infty}^{\infty} x(n\Delta) e^{-i2\pi fn\Delta}$$

Q: What is the difference between $X_s(f)$ and $X(f)$?

Impluse Train Modulation (Continue)

$$X_s(f + r/\Delta) = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi(f+r/\Delta)n\Delta} = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta - i2\pi rn} = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta} = X_s(f)$$

r: integer

Q: Meaning?

$$X_s(f) = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta}$$

Multiplying both sides of equation by $e^{-i2\pi fr\Delta}$ and integrating w.r.t f

$$\int_{-1/2\Delta}^{1/2\Delta} X_s(f)e^{-i2\pi fr\Delta}df = \int_{-1/2\Delta}^{1/2\Delta}\left[\sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta}\right]e^{-i2\pi fr\Delta}df = \sum_{n=-\infty}^{\infty}\int_{-1/2\Delta}^{1/2\Delta}\left[x(n\Delta)e^{-i2\pi fn\Delta}\right]e^{-i2\pi fr\Delta}df$$

$$= \sum_{n=-\infty}^{\infty} x(n\Delta)\int_{-1/2\Delta}^{1/2\Delta}\left[e^{-i2\pi f(n-r)\Delta}\right]df = x(r\Delta)\frac{1}{\Delta}$$

Use L'Hôpital's rule

$$f_S = \frac{1}{\Delta}$$

$$X_s(f) = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta} \qquad x(n\Delta) = \Delta\int_{-1/2\Delta}^{1/2\Delta} X_s(f)e^{i2\pi fn\Delta}df$$

Q: What are these equation?

Impluse Train Modulation (Continue) - Proof

Supplement

$$\sum_{n=-\infty}^{\infty} x(n\Delta) \int_{-1/2\Delta}^{1/2\Delta} \left[e^{-j2\pi f(n-n)\Delta} \right] df = \sum_{n=-\infty}^{\infty} x(n\Delta) \left[\frac{1}{-j2\pi(n-n)\Delta} e^{-j2\pi f(n-n)\Delta} \right]_{-1/2\Delta}^{1/2\Delta}$$

$$= \sum_{n=-\infty}^{\infty} x(n\Delta) \frac{1}{-j2\pi(n-n)\Delta} \left[e^{-\pi(n-n)} - e^{\pi(n-n)} \right]$$

$$= \sum_{n=-\infty}^{\infty} x(n\Delta) \frac{1}{\pi(n-n)\Delta} \left[e^{-\pi(n-n)} - e^{\pi(n-n)} \right] / 2j$$

$$= \sum_{n=-\infty}^{\infty} x(n\Delta) \frac{1}{\pi(n-n)\Delta} \sin(\pi(n-n))$$

$$= \sum_{n=-\infty}^{\infty} x(n\Delta) \frac{1}{\Delta} \cdot \frac{\sin(\pi(n-n))}{\pi(n-n)}$$

$$= x(n\Delta) \frac{1}{\Delta}$$

- $\sin\theta = \frac{1}{2j}(e^{j\theta} - e^{-j\theta})$ ← Euler's formula.
- $\lim_{x \to 0} \frac{\sin x}{x} = 1$ L'Hôpital's rule
- $\frac{\sin(\pi(n-n))}{\pi(n-n)} = 1$ When $n = n$

  $= 0$ otherwise.

Link Between Fourier Transform of a Discrete Sequence and Continuous Signal

Fourier coefficients

$$x(t) = \sum_{n=-\infty}^{\infty} c_n e^{i2\pi nt/T_p} \qquad c_n = \frac{1}{T_p}\int_0^{T_p} x(t)e^{-i2\pi nt/T_p}dt$$

Impluse train

$$i(t) = \sum_{m=-\infty}^{\infty} \delta(t - m\Delta) \qquad c_n = \frac{1}{\Delta}\int_{-\Delta/2}^{\Delta/2} \sum_{m=-\infty}^{\infty} \delta(t - m\Delta)\, e^{-i2\pi nt/\Delta}dt == \frac{1}{\Delta}\int_{-\Delta/2}^{\Delta/2} \delta(t)e^{-i2\pi nt/\Delta}dt = \frac{1}{\Delta}$$

Fourier Transform of the impulse train

$$I(f) = F\big(i(t)\big) = \int_{-\infty}^{\infty} i(t)e^{-i2\pi ft}dt = \int_{-\infty}^{\infty} \sum_{n=-\infty}^{\infty} c_n e^{\frac{i2\pi nt}{T_p}} \cdot e^{-i2\pi ft}dt$$

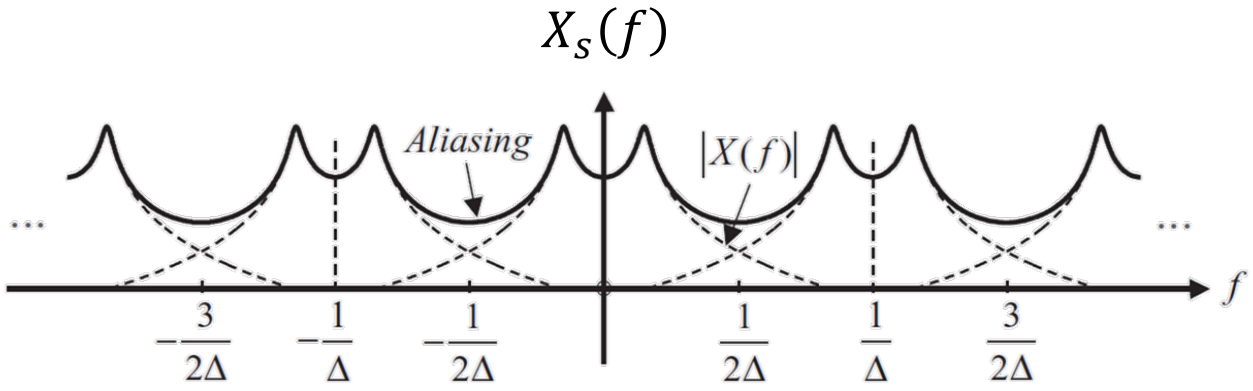
$$= \int_{-\infty}^{\infty} \sum_{n=-\infty}^{\infty} \frac{1}{\Delta} e^{\frac{i2\pi nt}{\Delta}} \cdot e^{-i2\pi ft}dt = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} \int_{-\infty}^{\infty} e^{\frac{i2\pi nt}{\Delta}} \cdot e^{-i2\pi ft}\, dt = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} \int_{-\infty}^{\infty} e^{-i2\pi\left(f - \frac{n}{\Delta}\right)t}dt = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} \delta(f - \frac{n}{\Delta})$$

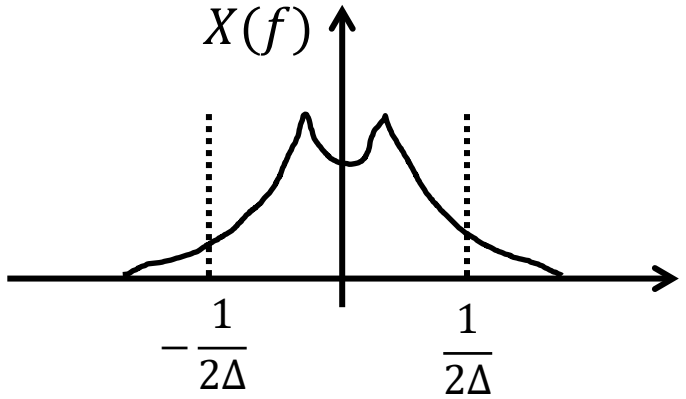
Link Between Fourier Transform of a Discrete Sequence and Continuous Signal (Continue)

Fourier Transform of a discrete sequence, $x_s(t) = x(t)i(t)$

$$X_s(f) = I(f) * X(f) = \int_{-\infty}^{\infty} I(g)X(f-g)dg = \int_{-\infty}^{\infty} \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} \delta\left(g - \frac{n}{\Delta}\right)X(f-g)dg = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty}\int_{-\infty}^{\infty} \delta\left(g - \frac{n}{\Delta}\right)X(f-g)dg$$

$$= \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right) = \frac{1}{\Delta}\left(\ldots + X\left(f - \frac{2}{\Delta}\right) + X\left(f - \frac{1}{\Delta}\right) + X(f) + X\left(f + \frac{1}{\Delta}\right) + \ \ldots\right)$$



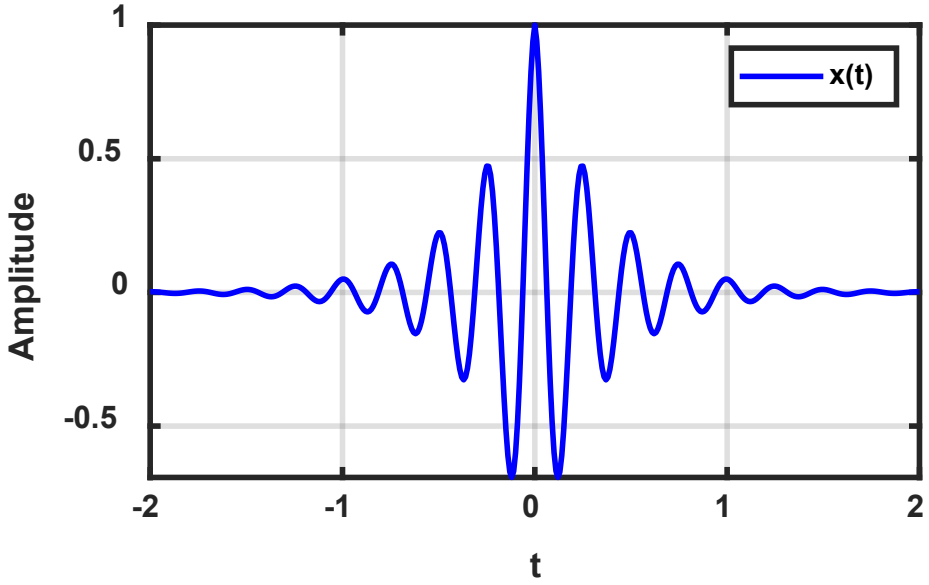


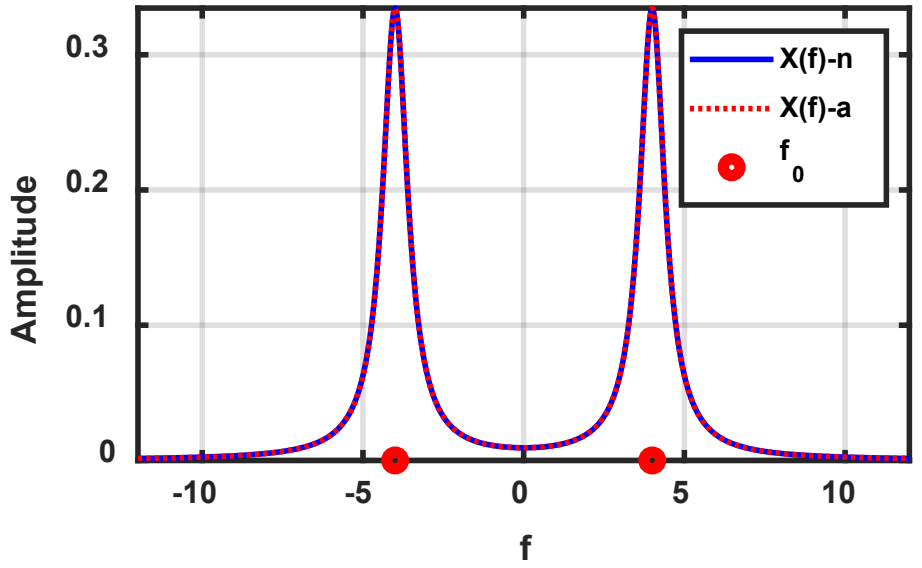
Q: What happened if Δ goes to a zero?

Example: Fourier Transform of a Discrete Sequence and Continuous Signal

$$x(t) = e^{-a|t|}cos2\pi f_0 t \quad \Rightarrow \quad X(f) = \frac{a}{a^2 + [2\pi(f - f_0)]^2} + \frac{a}{a^2 + [2\pi(f + f_0)]^2}$$

$$X(f) = \int_{-\infty}^{\infty} x(t)e^{-i2\pi f t}dt$$





Note that –n and –a indicate numerical and analytical integration for Fourier transformation.

Example: Fourier Transform of a Discrete Sequence and Continuous Signal (Continue)

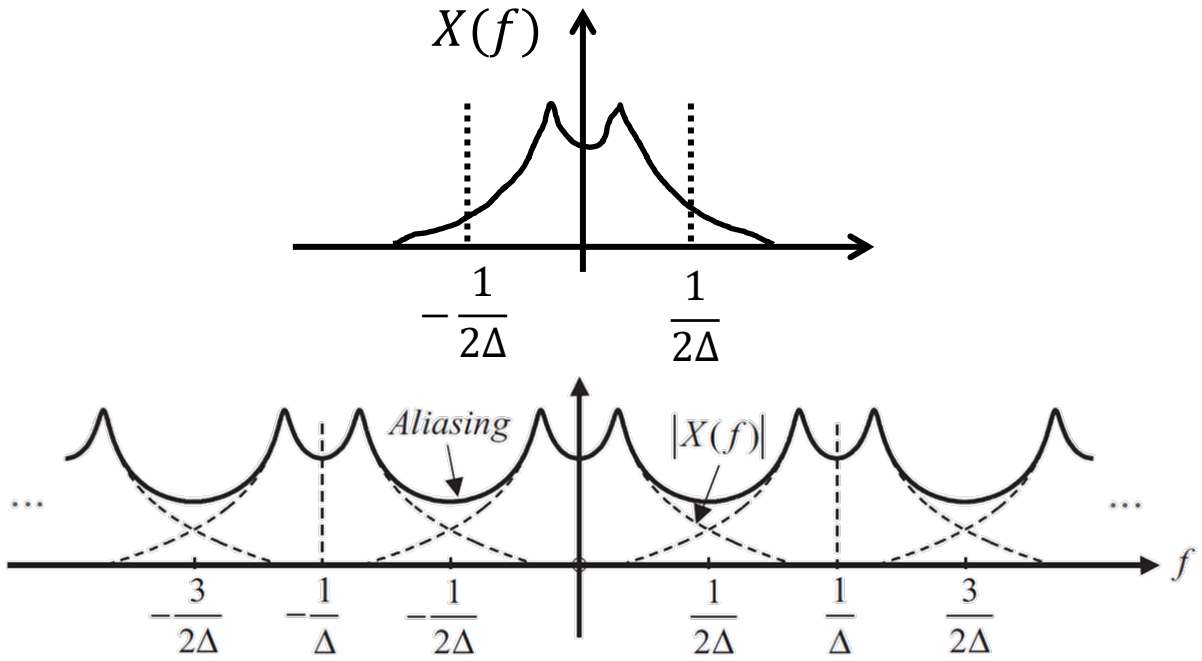
$$x(t) = e^{-a|t|}cos2\pi f_0 t \quad \Rightarrow \quad X(f) = \frac{a}{a^2 + [2\pi(f - f_0)]^2} + \frac{a}{a^2 + [2\pi(f + f_0)]^2}$$

$$X(f) = \int_{-\infty}^{\infty} x(t)e^{-i2\pi ft}dt$$

$$x_s(t) = x(t)i(t) \quad \Rightarrow \quad X_s(f) = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta}$$

$$X_s(f) = I(f) * X(f) = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right)$$

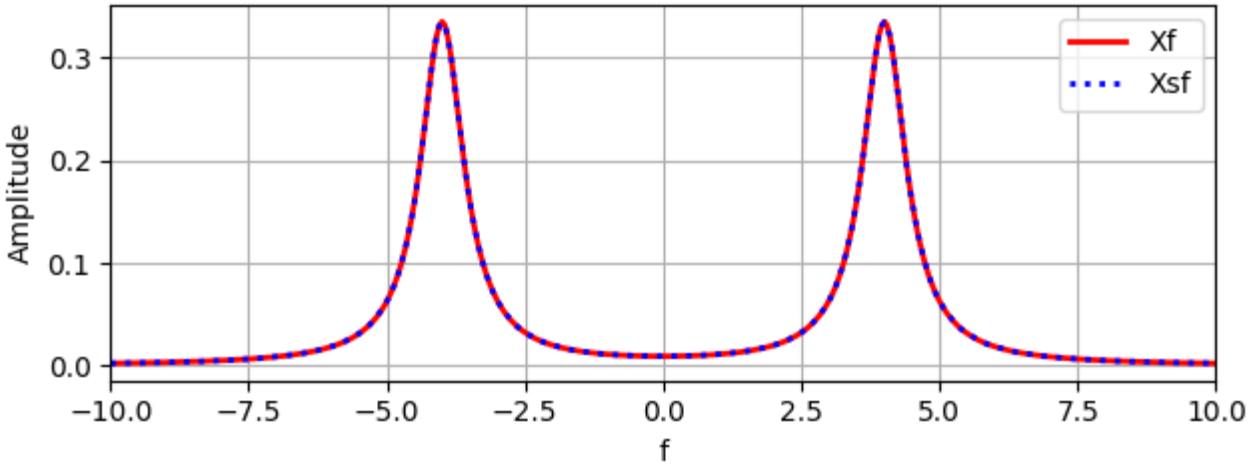
$$= \frac{1}{\Delta}\left(\ldots + X\left(f - \frac{1}{\Delta}\right) + X(f) + X\left(f + \frac{1}{\Delta}\right) + \ldots\right)$$

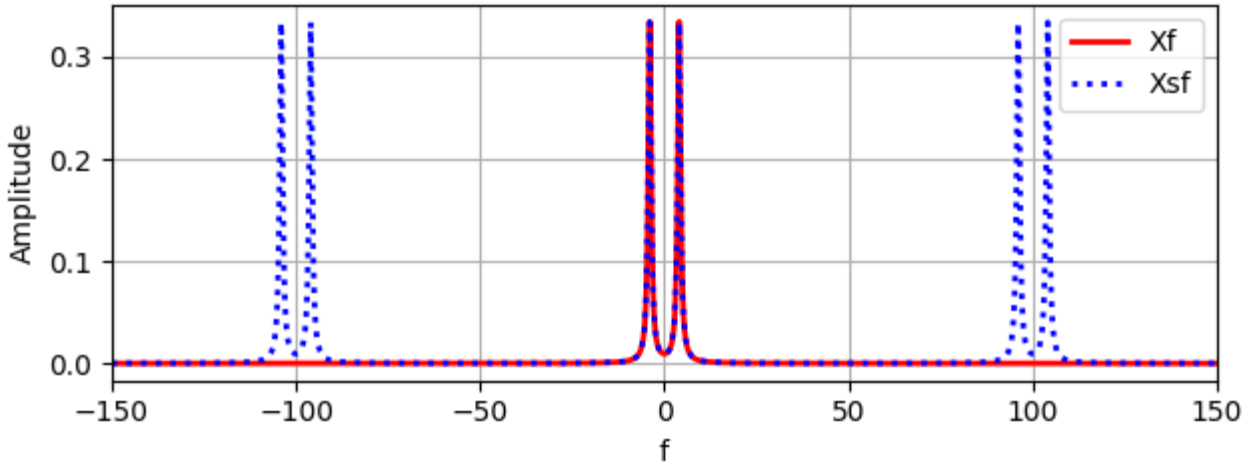


Comparison of Fourier Transform of a Discrete Sequence and Continuous Signal (Damped Symmetrically Oscillating Function)

The sampling frequncy is 100 Hz.

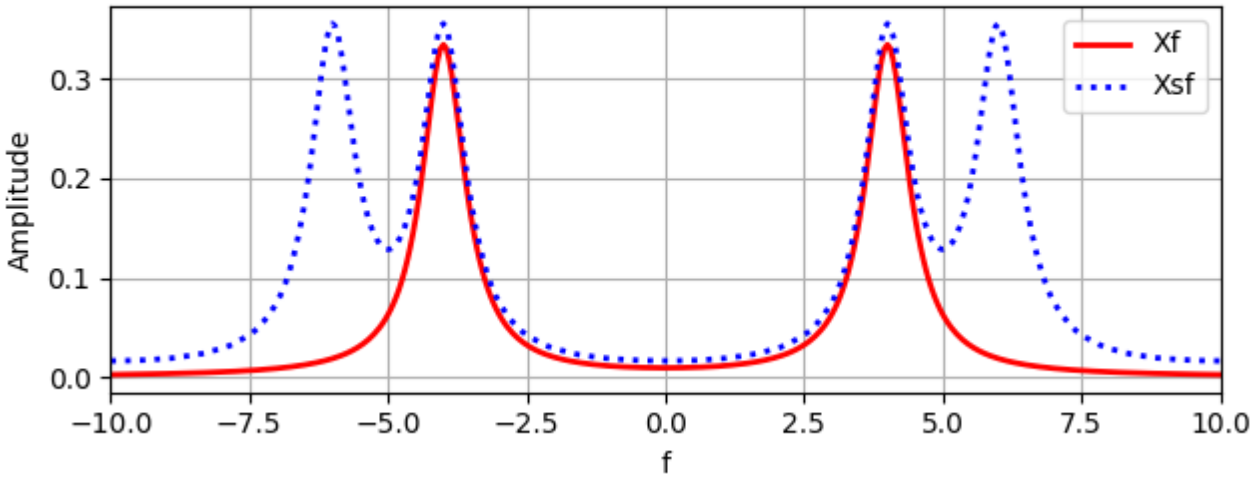
No aliasing

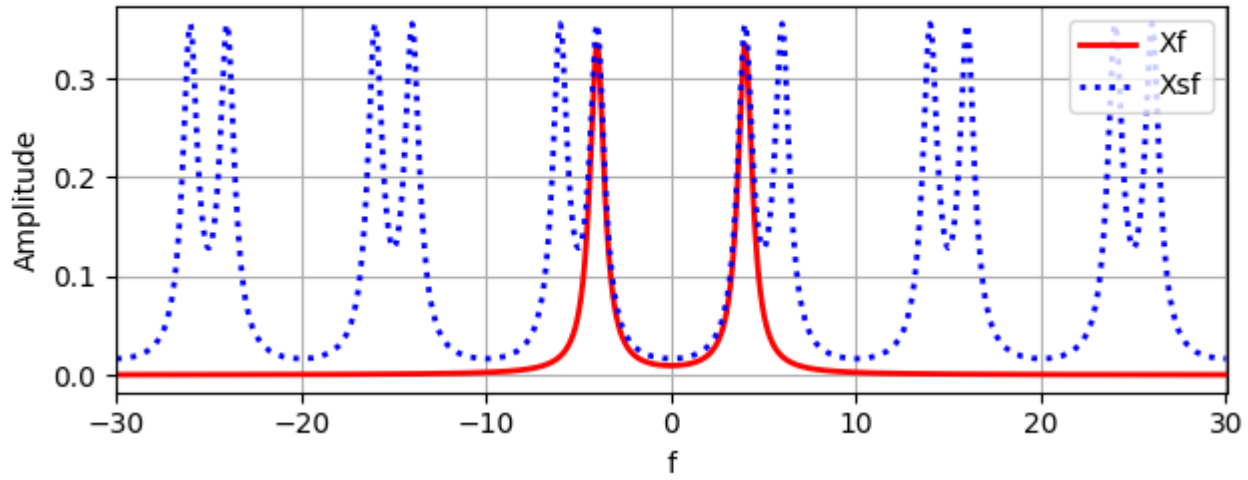




The sampling frequncy is 10 Hz.

Aliasing





Example: Continuous Fourier Transform of a Discrete Signal

```python
a = 3
f0 = 4

# Define functions
x1 = lambda t: np.exp(-a * np.abs(t))
x2 = lambda t: np.cos(2 * np.pi * f0 * t)
x = lambda t: x1(t) * x2(t)

# Define Xf and Xsf for a given fs and t
def define_functions(fs, t):
    xn = x(t) # t is n * delta
    Xf = lambda f: a / (a**2 + (2 * np.pi * (f - f0))**2) + a / (a**2 + (2 * np.pi * (f + f0))**2)
    Xsf = lambda f: (1/fs) * np.sum(xn * np.exp(-1j * 2 * np.pi * f * t))
    return Xf, Xsf
```

```python
# Plot for fs = 100: low sampling frequency
fs = 100
dt = 1/fs
t = np.arange(-2, 2, dt)
plot(fs, t, 1)

# Plot for fs = 10: low sampling frequency
fs = 10
dt = 1/fs
t = np.arange(-2, 2, dt)
plot(fs, t, 2)
```

$$x(t) = e^{-a|t|} cos 2\pi f_0 t$$

$$X(f) = \frac{a}{a^2 + [2\pi(f - f_0)]^2} + \frac{a}{a^2 + [2\pi(f + f_0)]^2}$$

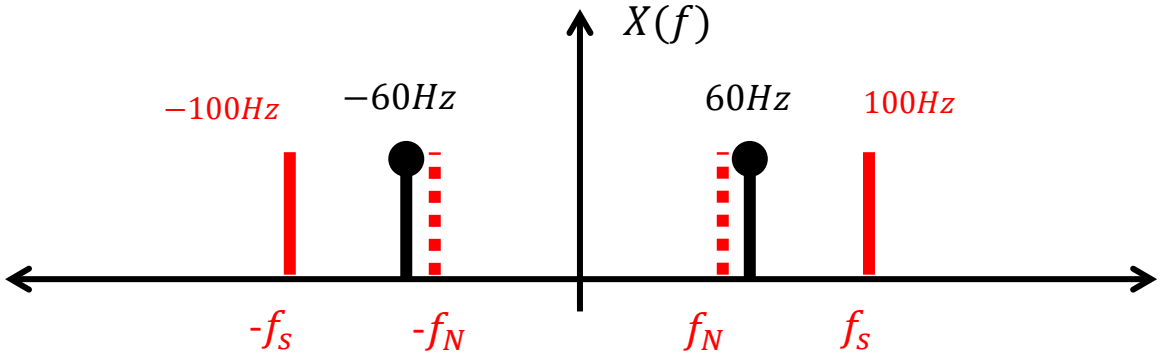
$$X_s(f) = \sum_{n=-\infty}^{\infty} x(n\Delta) e^{-i2\pi f n\Delta}$$

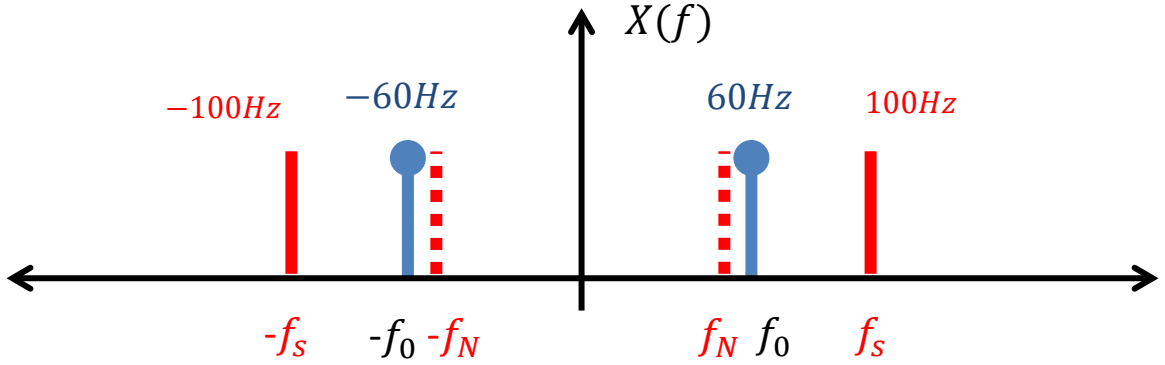
$$X_s(f) = I(f) * X(f) = \frac{1}{\Delta} \sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right)$$

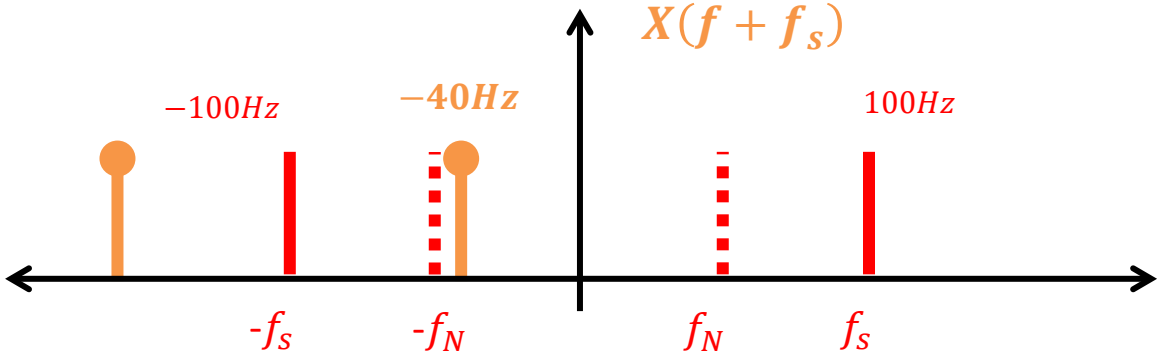
Example: Wagon-Wheel Effect 1

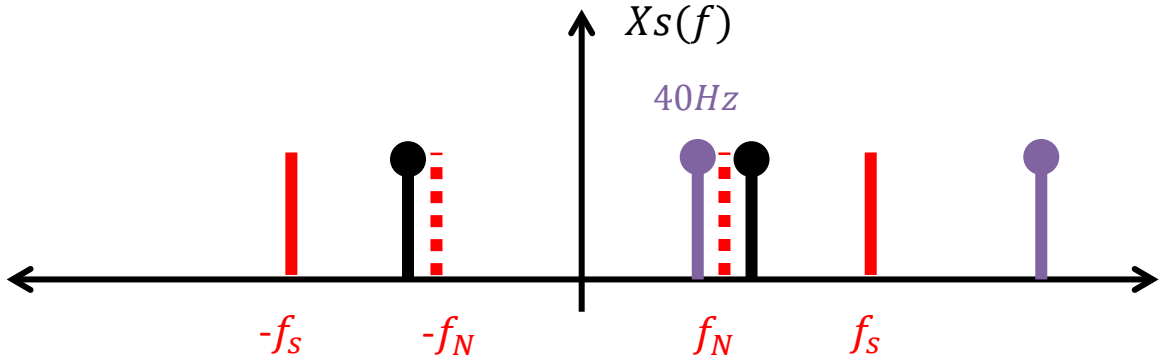
$$X_s(f) = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right) = f_s \sum_{n=-\infty}^{\infty} X(f - nf_s)$$

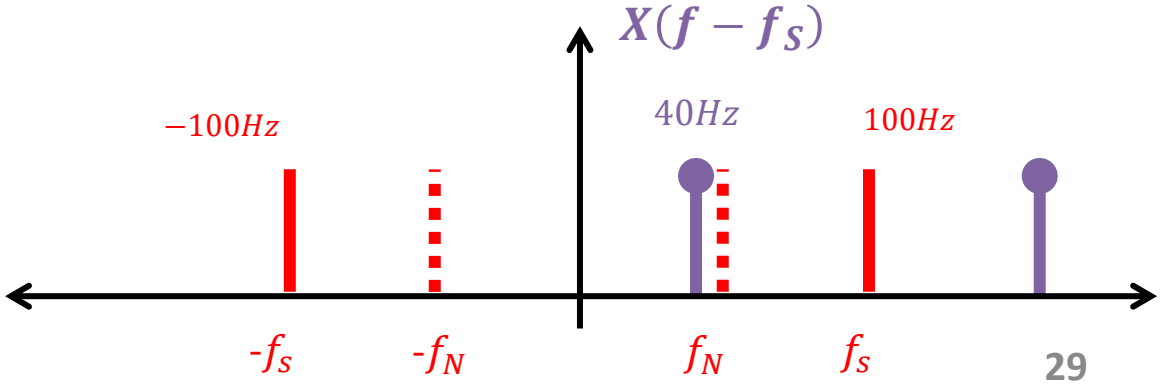
$$\frac{1}{\Delta}\left(\ldots + X\left(f - \frac{2}{\Delta}\right) + X\left(f - \frac{1}{\Delta}\right) + X(f) + X\left(f + \frac{1}{\Delta}\right) + \ldots\right)$$







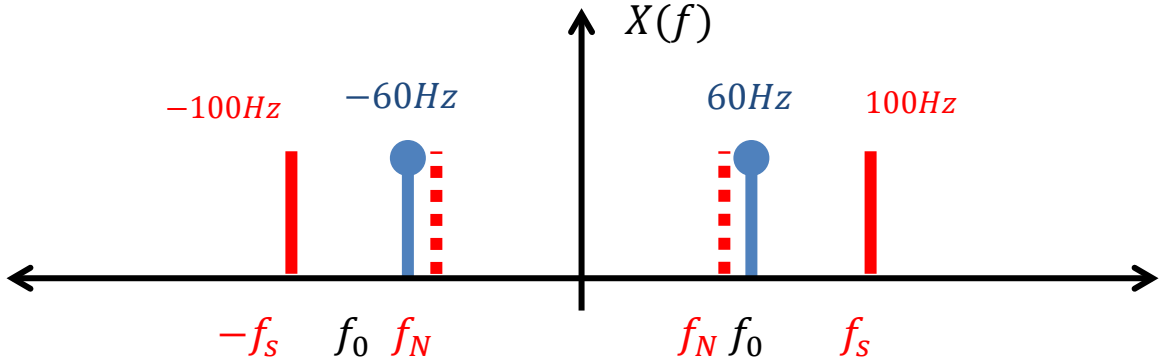


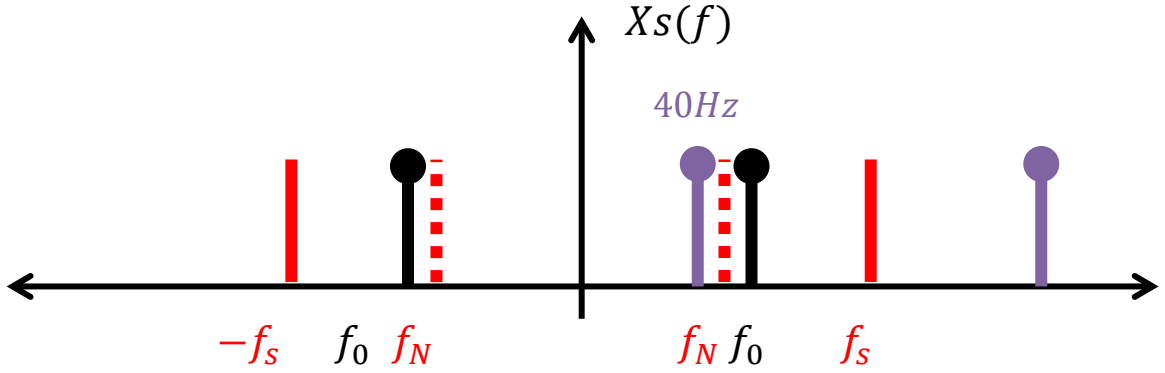


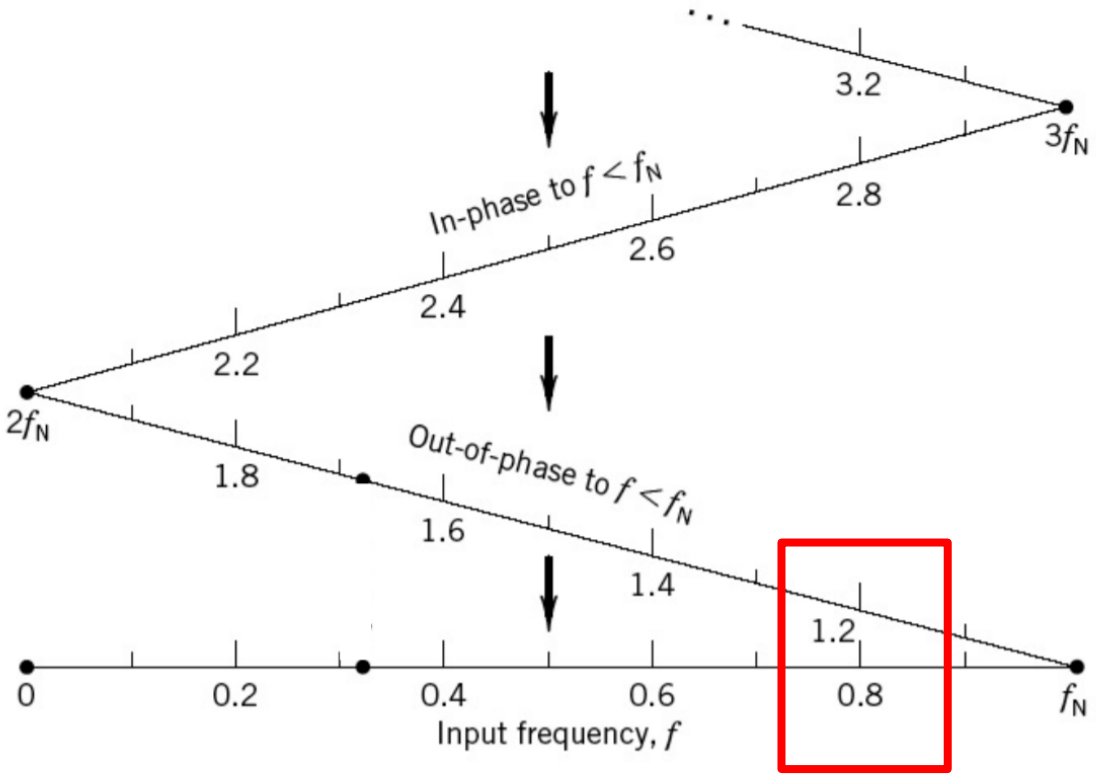
Example: Wagon-Wheel Effect 1

$$X_s(f) = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right) = f_s \sum_{n=-\infty}^{\infty} X(f - nf_s)$$

$$\frac{1}{\Delta}\left(\ldots + X\left(f - \frac{2}{\Delta}\right) + X\left(f - \frac{1}{\Delta}\right) + X(f) + X\left(f + \frac{1}{\Delta}\right) + \ldots\right)$$





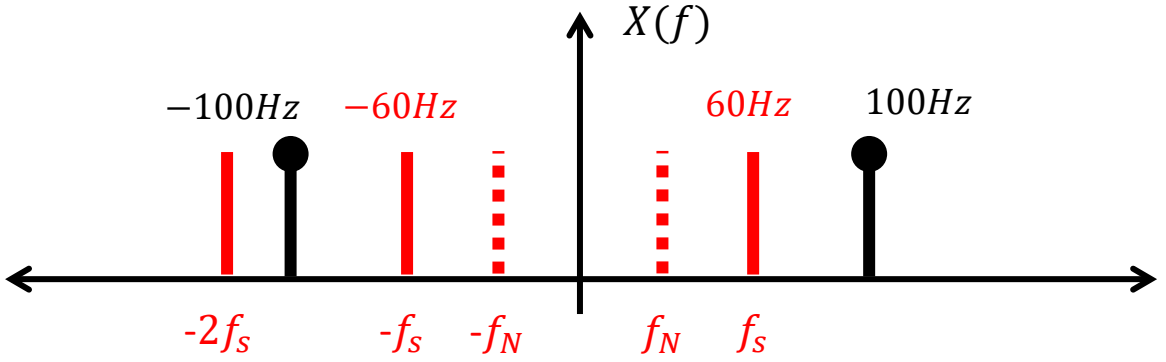


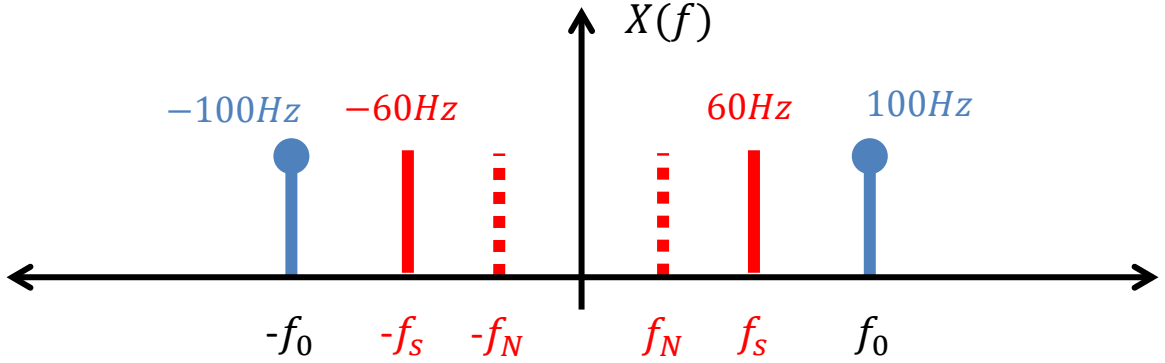
- f_0 = 60 Hz
- f_s = 100 Hz
- f_N = 50 Hz
- Frequency after aliasing: 0.8 f_N = 40 Hz

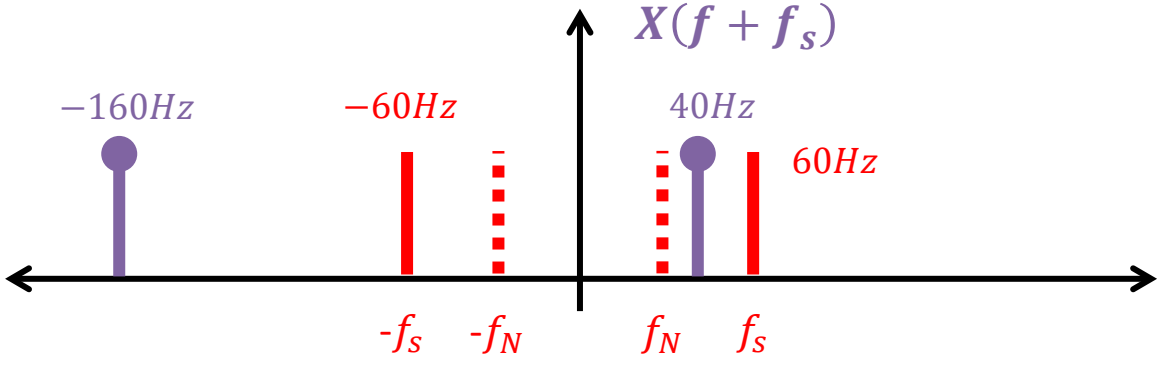
Example: Wagon-Wheel Effect 2

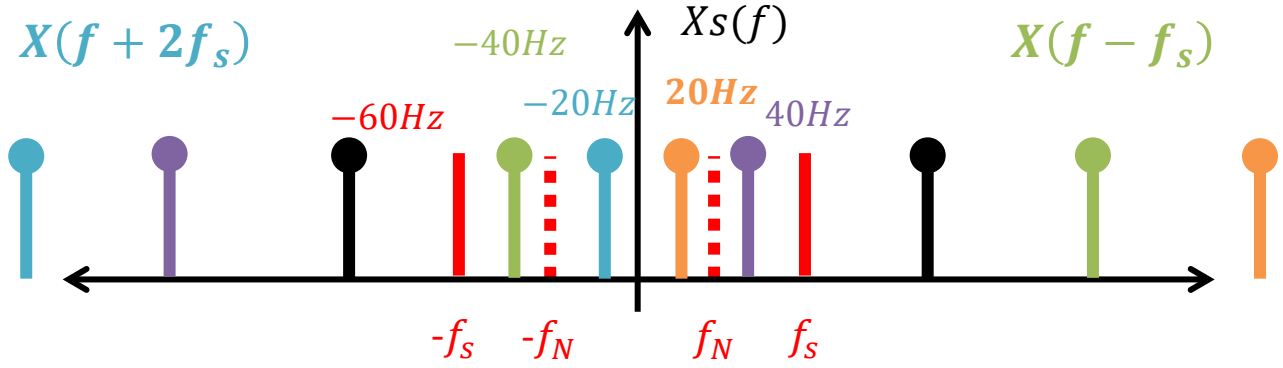
$$X_s(f) = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right) = f_s \sum_{n=-\infty}^{\infty} X(f - nf_s)$$

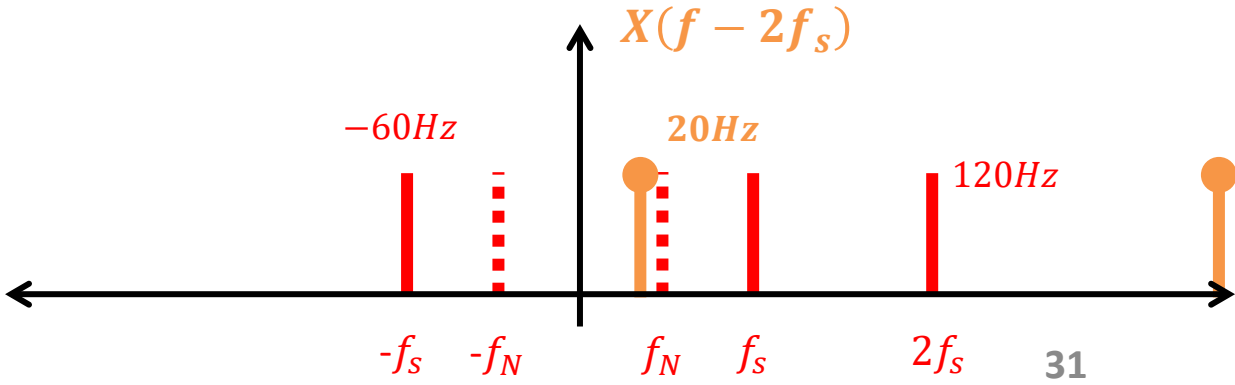
$$\frac{1}{\Delta}\left(\ldots + X\left(f - \frac{2}{\Delta}\right) + X\left(f - \frac{1}{\Delta}\right) + X(f) + X\left(f + \frac{1}{\Delta}\right) + \ldots\right)$$







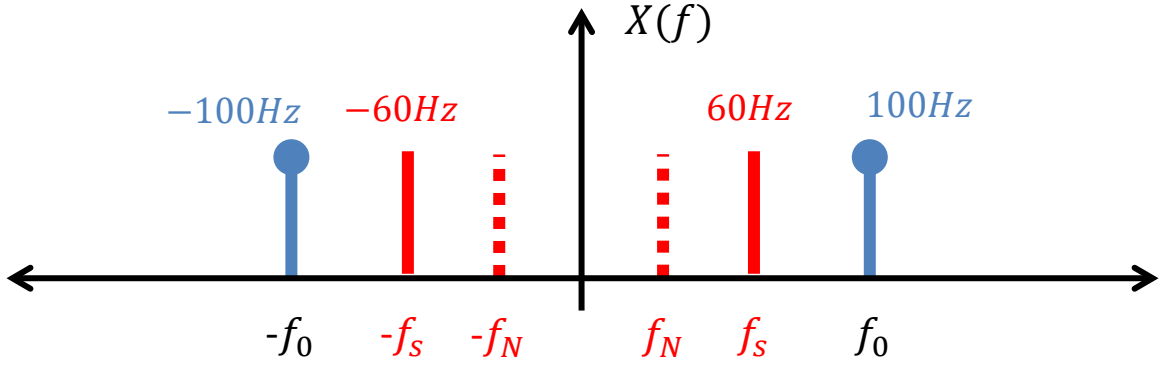


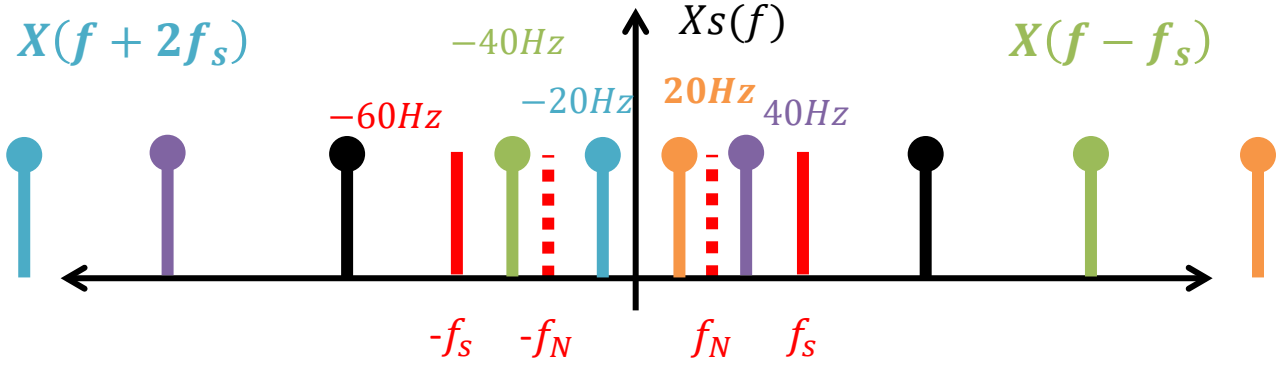


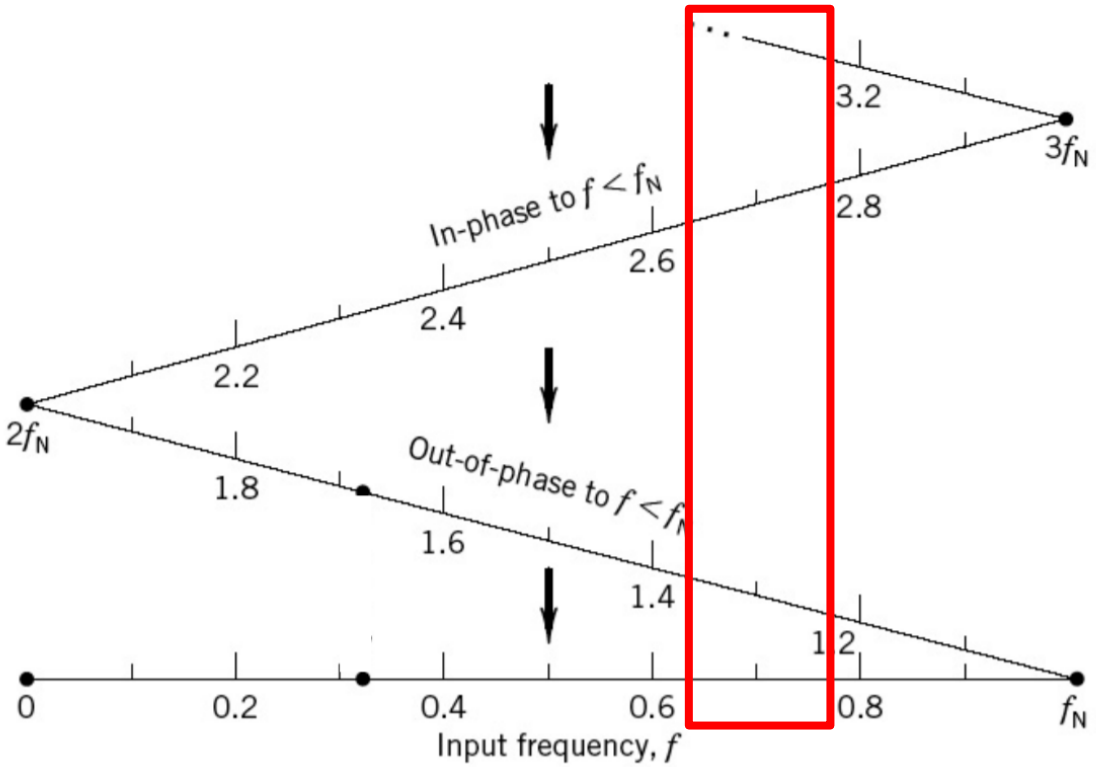
Example: Wagon-Wheel Effect 2

$$X_s(f) = \frac{1}{\Delta}\sum_{n=-\infty}^{\infty} X\left(f - \frac{n}{\Delta}\right) = f_s \sum_{n=-\infty}^{\infty} X(f - nf_s)$$

$$\frac{1}{\Delta}\left(\ldots + X\left(f - \frac{2}{\Delta}\right) + X\left(f - \frac{1}{\Delta}\right) + X(f) + X\left(f + \frac{1}{\Delta}\right) + \ldots\right)$$

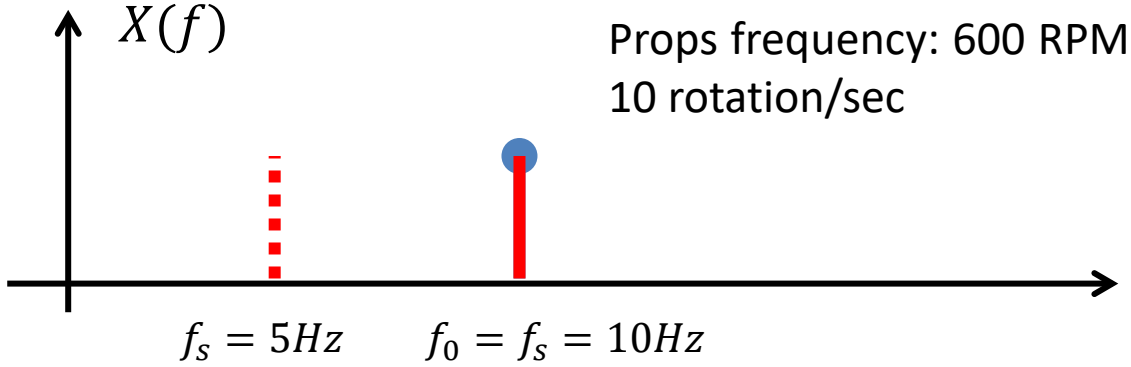




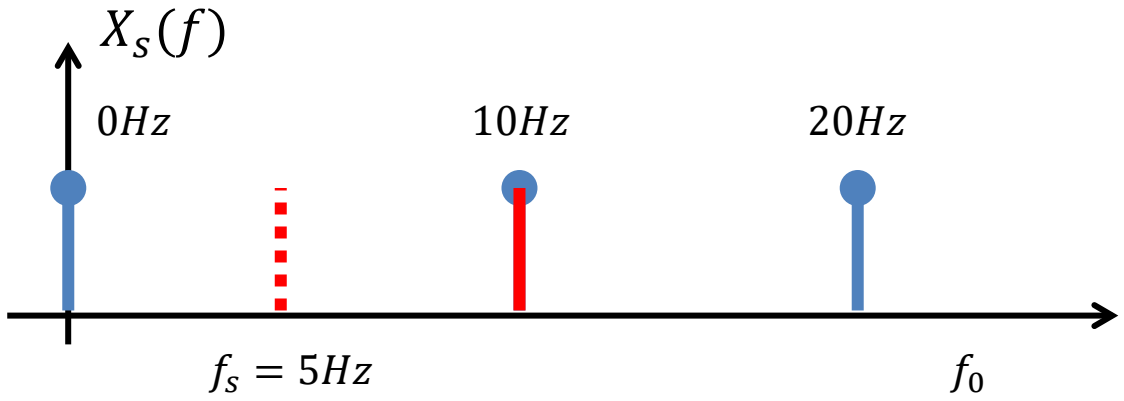


- f_0 = 100 Hz
- f_s = 60 Hz
- f_N = 30 Hz
- Frequency after aliasing: 10/3 f_N -> 2/3 f_N = 20 Hz

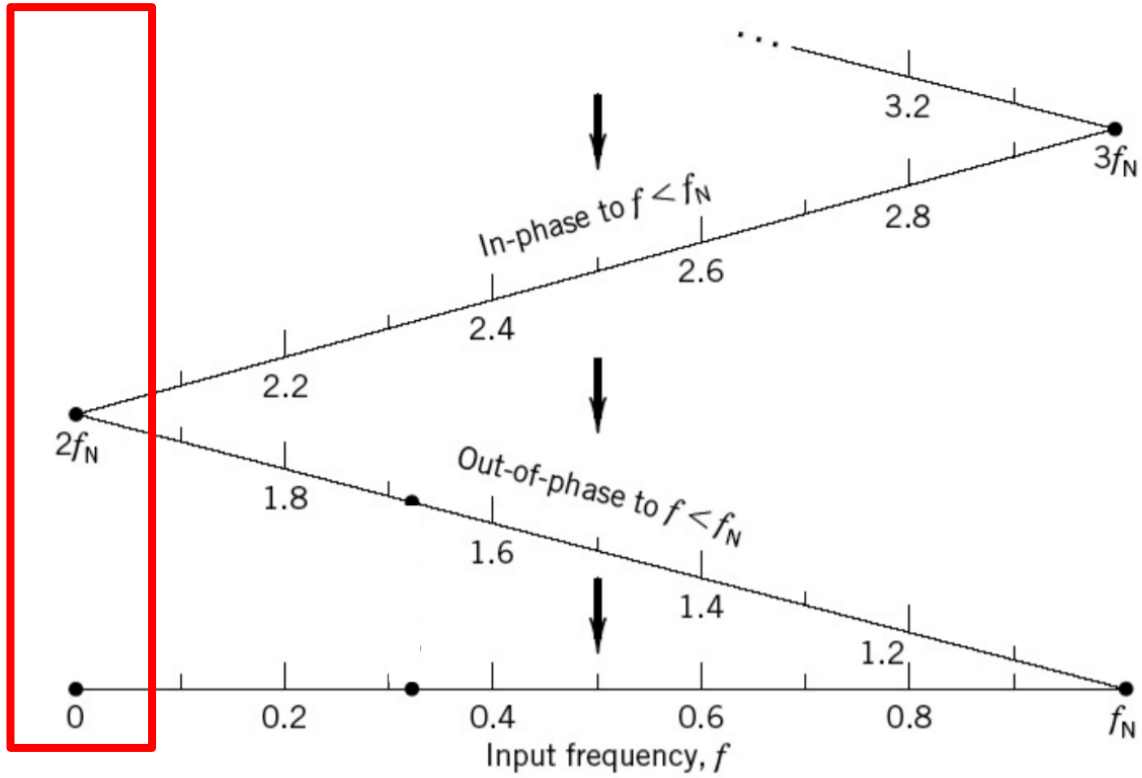
Revisit: Spaceship Helicopter



Props frequency: 600 RPM = 10 rotation/sec



The frequency is wrapped around to zero frequency!!



- f_0 = 10 Hz
- f_s = 10 Hz
- f_N = 5 Hz
- Frequency after aliasing: 0 Hz

Discrete Fourier Transform

So far, we have considered sequences that run over the range $-\infty < n < \infty$(n integer). For the special case where the sequence is of finite length (i.e. non-zero for a finite number of values) an alternative Fourier representation is possible called the **discrete Fourier transform (DFT)**.

It turns out that the DFT is a Fourier representation of a finite length sequence and is itself a sequence rather than a continuous function of frequency, and it corresponds to samples, **equally spaced in frequency**, of the Fourier transform of the signal. The DFT is fundamental to many digital signal processing algorithms.

$$X_s(f) = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta}$$

Continuous in frequency

Repeated every $1/\Delta$

$$X(k) = \sum_{n=0}^{N-1} x(n)e^{-i\frac{2\pi}{N}nk}$$

DFT of a finite (sampled) sequence $x(n\Delta)$ $\quad f = \frac{k}{N\Delta}$

$X(k)$ is $X_s(f)$ evaluated at $f = \frac{k}{N\Delta}$ Hz (k integer)

$$x(n) = \frac{1}{N}\sum_{n=0}^{N-1} X(k)e^{i\frac{2\pi}{N}nk}$$

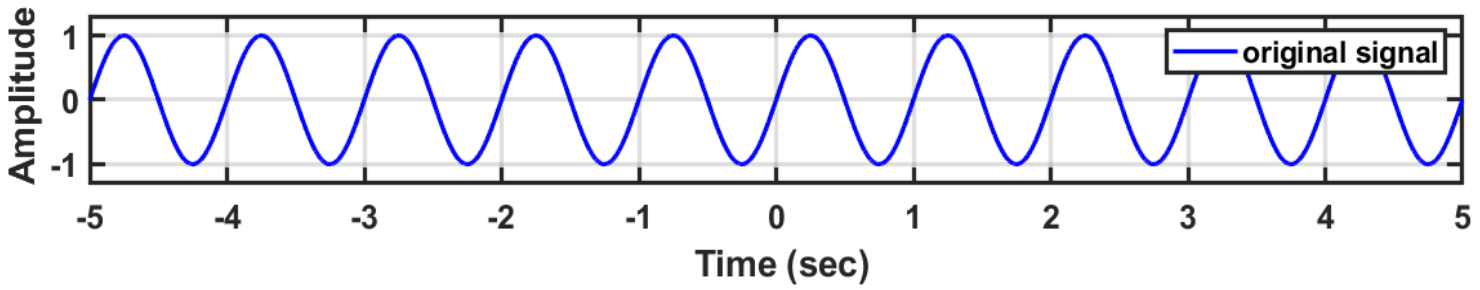
Inverse DFT

Q: How do we store frequency signals?

Discrete Fourier Transform (Continue)

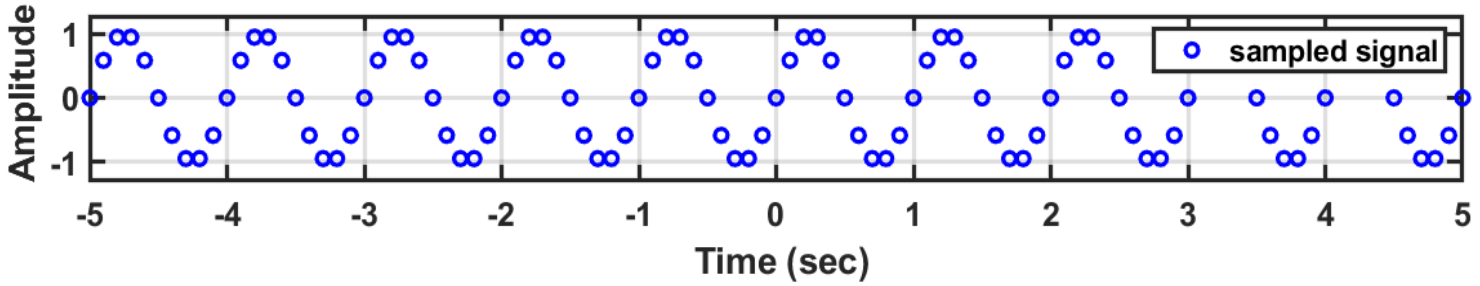
Fourier transform of a **continuous** time signal $x(t)$

$$X(f) = \int_{-\infty}^{\infty} x(t)e^{-i2\pi ft}dt$$



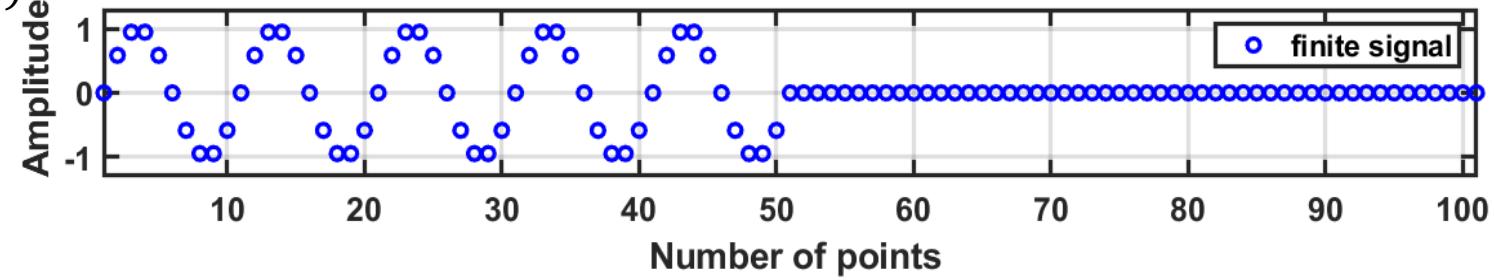
Fourier transform of a **discrete** sequence $x(n\Delta)$

$$X_s(f) = \sum_{n=-\infty}^{\infty} x(n\Delta)e^{-i2\pi fn\Delta}$$



Fourier transform of a **discrete finite** sequence $x(n)$

$$X_s(f) = \sum_{n=0}^{N-1} x(n)e^{-i2\pi fn\Delta}$$



Discrete Fourier transform of a **discrete finite** sequence $x(n)$

$$X(k) = \sum_{n=0}^{N-1} x(n)e^{-i\frac{2\pi}{N}nk}$$

$X(k)$ is $X_s(f)$ evaluated at $f = \frac{k}{N\Delta}$ Hz (k integer)

Tutorial

Fast Fourier Transform

A fast Fourier transform (FFT) is an algorithm that computes the discrete Fourier transform (DFT) of a sequence, or its inverse (IDFT). Fourier analysis converts a signal from its original domain (often time or space) to a representation in the frequency domain and vice versa. It manages to reduce the complexity of computing the DFT from $O(n^2)$, which arises if one simply applies the definition of DFT, to $O(n\ log\ n)$, where n is the data size.

$$X(k) = \sum_{n=0}^{N-1} x(n) e^{-i\frac{2\pi}{N}nk}$$

FFT in Matlab

fft

Fast Fourier transform

Syntax

```
Y = fft(X)
Y = fft(X,n)
Y = fft(X,n,dim)
```

Y = fft(X) computes the discrete Fourier transform (DFT) of X using a fast Fourier transform (FFT) algorithm

Y = fft(X,n) returns the n-point DFT. If no value is specified, Y is the same size as X.

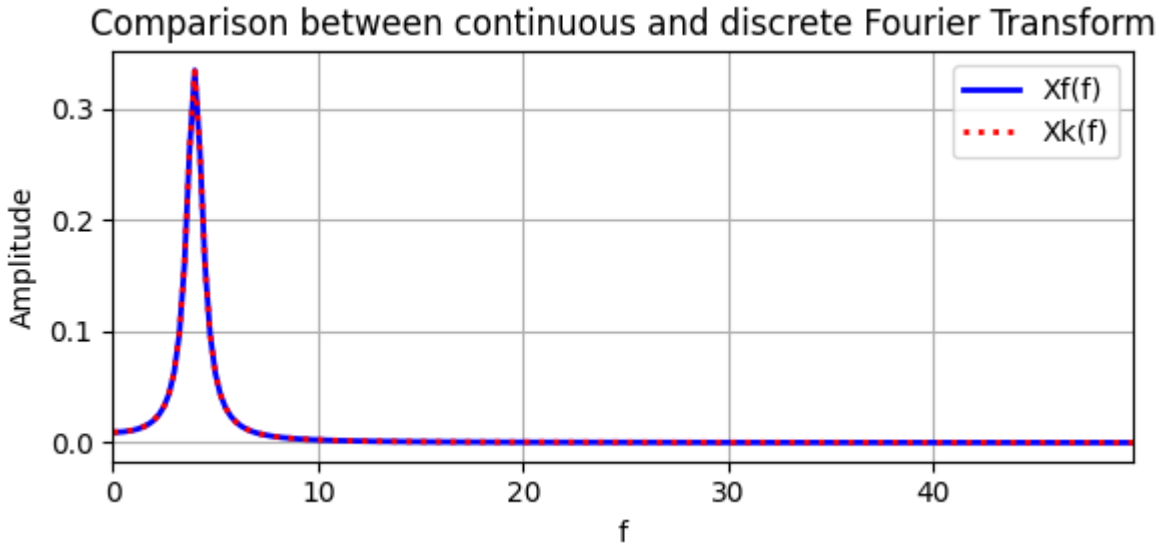
Example: Discrete Fourier Transform

```
a = 3
f0 = 4

# Define functions
x1 = lambda t: np.exp(-a * np.abs(t))
x2 = lambda t: np.cos(2 * np.pi * f0 * t)
x = lambda t: x1(t) * x2(t)

fs = 100
dt = 1/fs
t = np.arange(-2, 2, dt)
xn = x(t) # sampling

# Define Xf function
Xf = lambda f: a / (a**2 + (2 * np.pi * (f - f0))**2) \
     + a / (a**2 + (2 * np.pi * (f + f0))**2)

# Perform FFT
nfft = len(xn)
Xk = np.fft.fft(xn, nfft)

f = 1 / (nfft * dt) * np.arange(nfft//2)
```

$X(k)$ is $X_s(f)$ evaluated at $f = \frac{k}{N\Delta}$ Hz (k integer)

$$X_s(f) = \sum_{n=0}^{N-1} x(n) e^{-i2\pi f n \Delta}$$

$$X(k) = \sum_{n=0}^{N-1} x(n) e^{-i\frac{2\pi}{N} n k}$$



Zero-Padding

$$X(k) = \sum_{n=0}^{N-1} x(n) e^{-i\frac{2\pi}{N}nk}$$

Y = fft(X, n) returns the n-point DFT. If no value is specified, Y is the same size as X.

- If X is a vector and the length of X is less than n, then X is padded with trailing zeros to length n.
- If X is a vector and the length of X is greater than n, then X is truncated to length n.

$X(k)$ is $X_s(f)$ evaluated at $f = \frac{k}{N\Delta}$ Hz

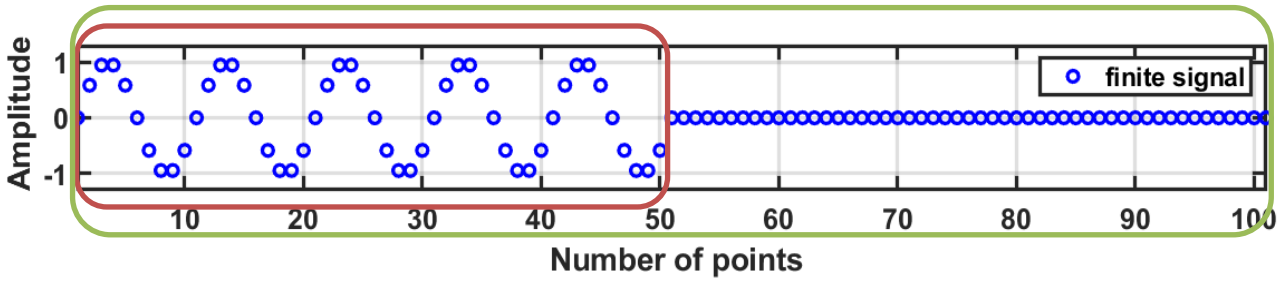
frequency spacing $\frac{1}{N\Delta}$ Hz

fft

Fast Fourier transform

Syntax

```
Y = fft(X)
Y = fft(X,n)
Y = fft(X,n,dim)
```



Zero-Padding (Continue)

$$\hat{x}(t) = x(n) \quad \text{for } 0 \le n \le N-1$$
$$= 0 \qquad \text{for } N \le n \le L-1$$

$$\hat{X}(k) = \sum_{n=0}^{L-1} \hat{x}(n) e^{-i\frac{2\pi}{L}nk} = \sum_{n=0}^{N-1} x(n) e^{-i\frac{2\pi}{L}nk}$$

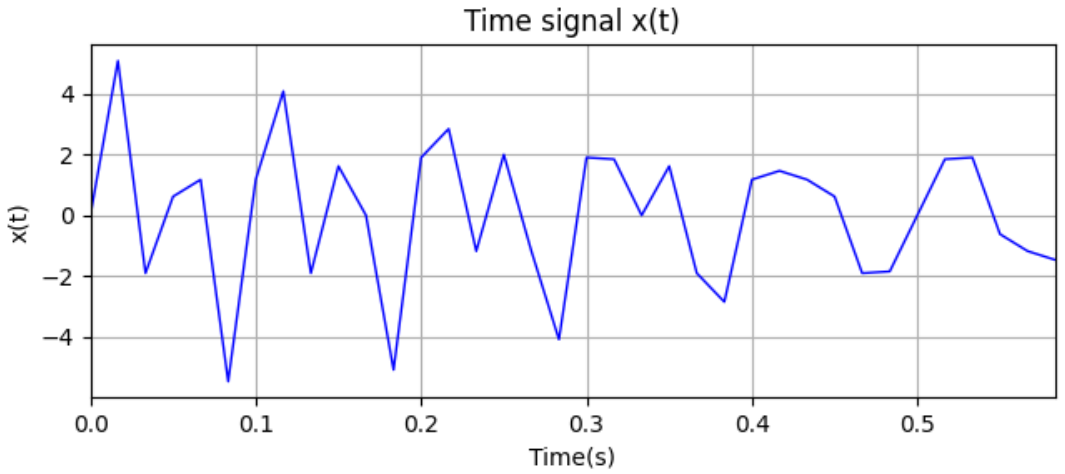
- "Finer" spacing in the frequency domain. However, the zero padding does not increase the "true" resolution.
- Vibration problems, this can be used to obtain the fine detail near resonances

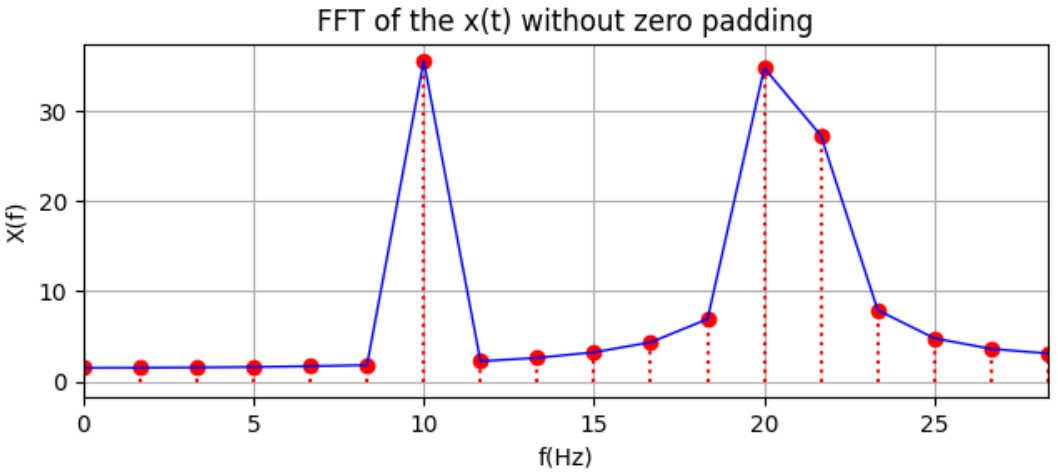
See the slide of "effect of data truncation"

Example: Zero-Padding Effect (Does not Increase a True Resolution)

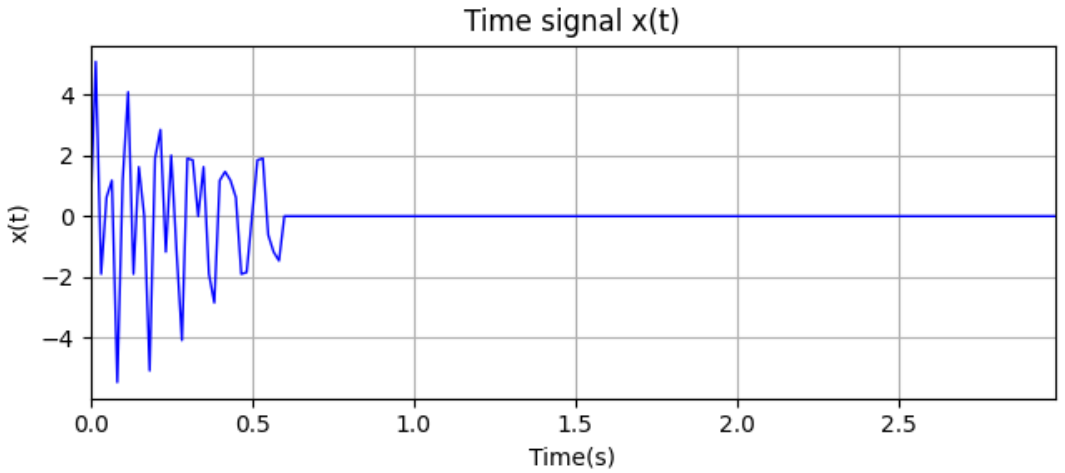
Black dotted lines indicate 10, 20, and 21 Hz

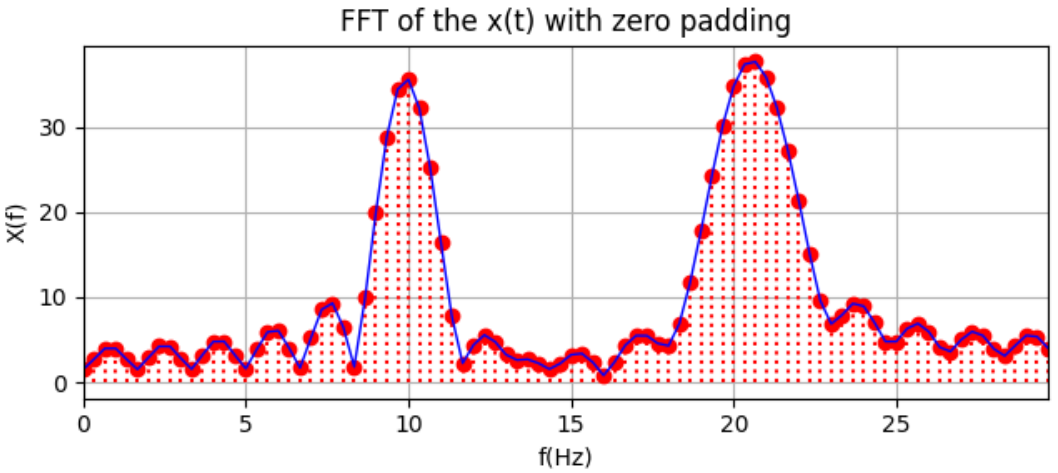
$T = 0.6$



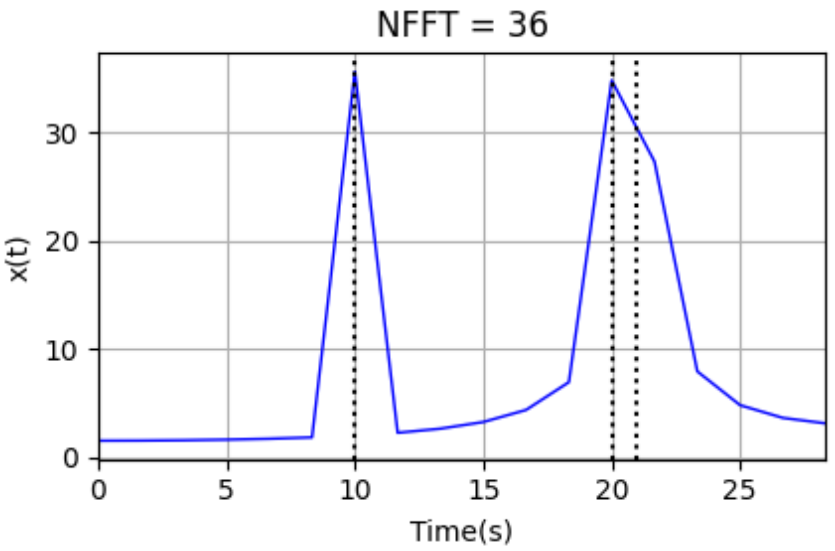


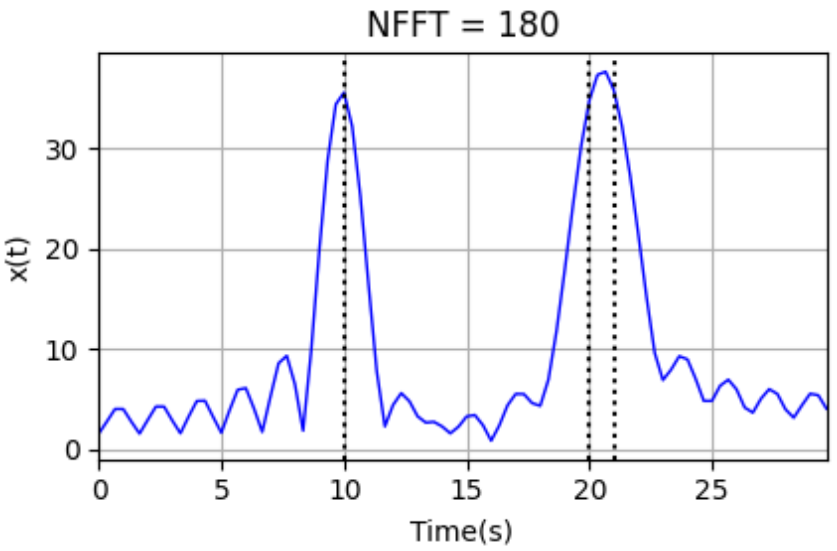
$T = 0.6$
+ zero padding
(2.4 s)

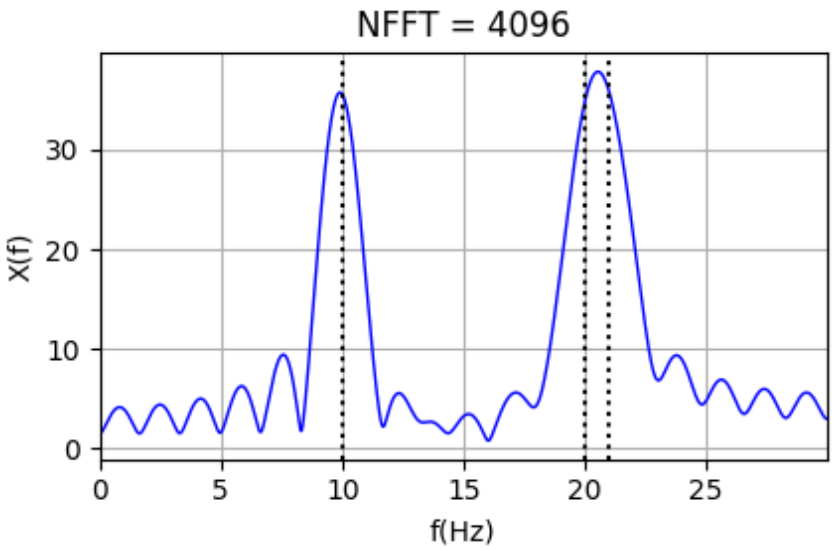




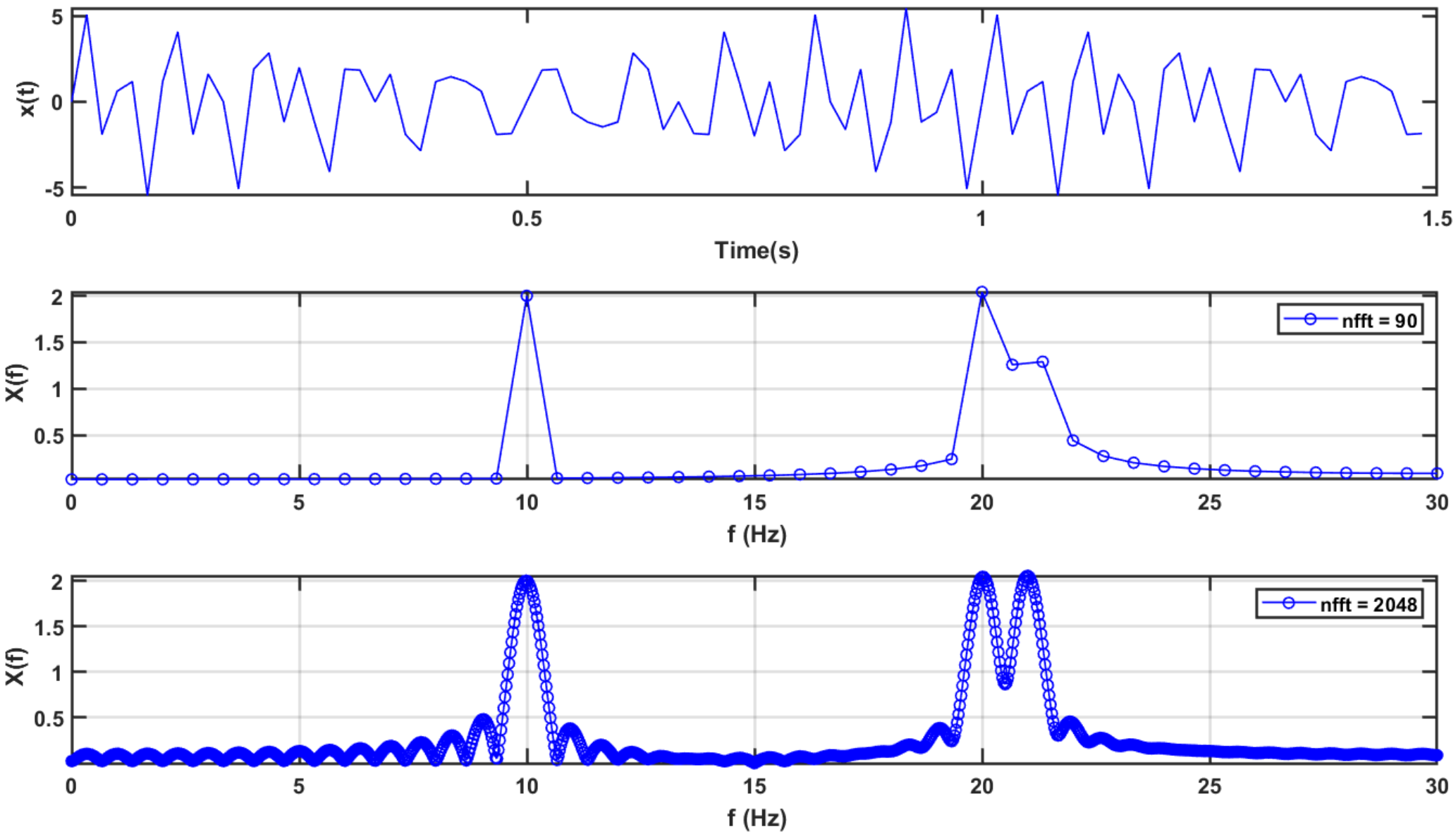
Comparison of Frequency Signals Depending on the number of FFT







Example: Zero-Padding Effect (Still Useful for Obtaining the Detail)



Waveform VS FFT Frequency Resolution

Supplement

The "**waveform frequency resolution**" is the minimum spacing between two frequencies that can be resolved. The "**FFT resolution**" is the number of points in the spectrum, which is directly proportional to the number points used in the FFT. **It is possible to have extremely fine FFT resolution, yet not be able to resolve two coarsely separated frequencies.** It is also possible to have fine waveform frequency resolution but have the peak energy of the sinusoid spread throughout the entire spectrum (this is called FFT spectral leakage).

The waveform frequency resolution is defined by the following equation: $$\Delta R = \frac{1}{T}$$

where T is the time length of the signal with data. It's important to note here that you should not include any zero padding in this time! Only consider the actual data samples.

It's important to make the connection here that the discrete time Fourier transform (DTFT) or FFT operates on the data as if it were an infinite sequence with zeros on either side of the waveform. This is why the FFT has the distinctive sinc function shape at each frequency bin. You should recognize the waveform resolution equation 1/T is the same as the space between nulls of a sinc function.

The FFT resolution is defined by the following equation: $$\Delta R = \frac{f_s}{N_{fft}}$$

https://www.bitweenie.com/listings/fft-zero-padding