052600 VU Signal and Image Processing

Fourier Transform 2: Introduction to 2D Fourier Transform

Torsten Möller + Jana Kemnitz + Raphael Sahann

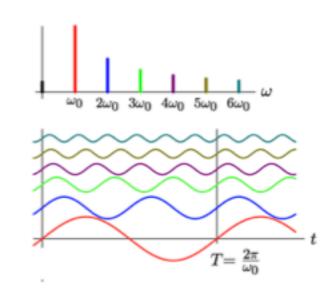
```
jana.kemnitz@univie.ac.at
raphael.sahann@univie.ac.at
```

vda.univie.ac.at/Teaching/SIP/18w

Recall - Fourier Series

- Fourier series is a way to represent a function as the sum of simple sine waves
- It decomposes any periodic signal into the sum of a set of simple sines and cosines (or complex exponentials)
- Recall Eulers formula:

$$e^{iy} = \cos(y) + i\sin(y)$$



Recall - Fourier Series

- Fourier Series is only applicable for periodic signals.
- Fourier Transform can be performed on aperiodic signals as well.
- It results in a function $F(\omega)$ that is continuous

Overview

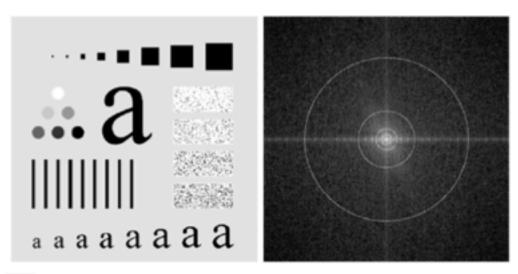
- Fourier Series (1D)
 - motivation
 - properties
 - examples
- Sampling & Impulse Train
- Fourier Transform (1D)
 - properties
 - convolution theorem
 - sampling in the Fourier space

All Fourier Transforms

	Spatial Domain	Frequency Domain
FT	$f(t) = \int_{-\infty}^{\infty} F(\omega) e^{j2\pi\omega t} dt$ continuous	$F(\omega) = \int_{-\infty}^{\infty} f(t) e^{-j2\pi\omega t} dt$ continuous
FS — Fourier Series	$f(t) = \sum_{n=-\infty}^{\infty} c_n e^{j\frac{2\pi n}{T}t}$ continuous + periodic	$c_n = \frac{1}{T} \int_{-T/2}^{T/2} f(t) e^{-j\frac{2\pi n}{T}t} dt$ discrete
DFT — Discrete FT	$f_n = \frac{1}{M} \sum_{m=0}^{M-1} F_m e^{j2\pi m n/M}$ discrete + periodic	$F_m = \sum_{n=0}^{M-1} f_n e^{-j2\pi mn/M}$ discrete + periodic
DTFT — Discrete Time FT	$f_n = \frac{1}{2\pi} \int_{-\pi}^{\pi} F(\omega) e^{j\omega n} d\omega$ discrete	$F(\omega) = \sum_{n=-\infty}^{\infty} f_n e^{-j\omega n} d\omega$ continuous + periodic

© Torsten Möller

Some intuition



a b

FIGURE 4.41 (a) Test pattern of size 688×688 pixels, and (b) its Fourier spectrum. The spectrum is double the image size due to padding but is shown in half size so that it fits in the page. The superimposed circles have radii equal to 10, 30, 60, 160, and 460 with respect to the full-size spectrum image. These radii enclose 87.0, 93.1, 95.7, 97.8, and 99.2% of the padded image power, respectively.

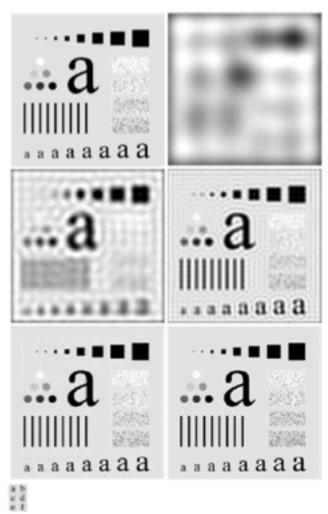
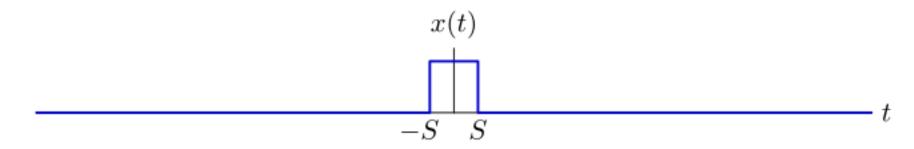


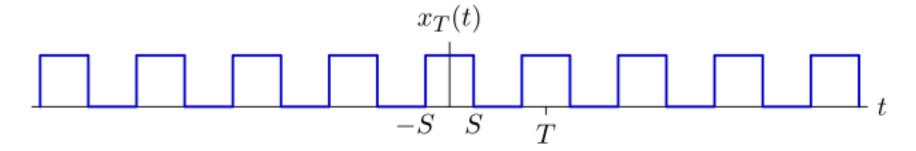
FIGURE 4.42 (a) Original image. (b)-(f) Results of filtering using ILPFs with cutoff frequencies set at radii values 10, 30, 60, 160, and 460, as shown in Fig. 4.41(b). The power removed by these filters was 13, 6.9, 4.3, 2.2, and 0.8% of the total, respectively.

An aperiodic signal can be thought of as periodic with infinite period.

Let x(t) represent an aperiodic signal.

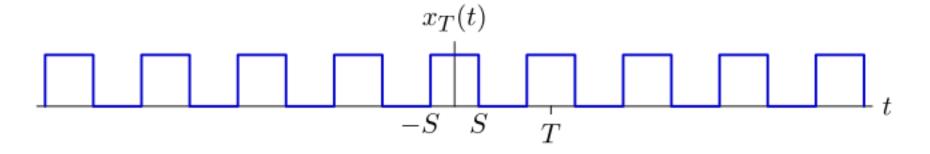


"Periodic extension":
$$x_T(t) = \sum_{k=-\infty}^{\infty} x(t+kT)$$

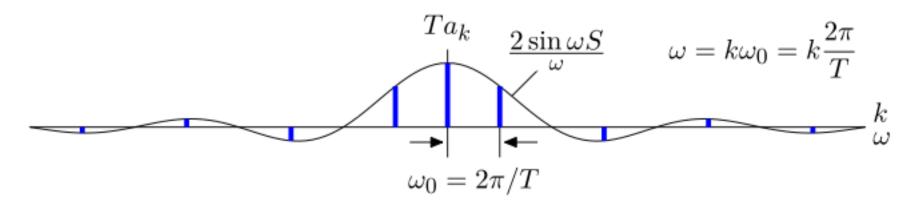


Then
$$x(t) = \lim_{T \to \infty} x_T(t)$$
. © Prof. Dennis Freeman, MIT, 2011

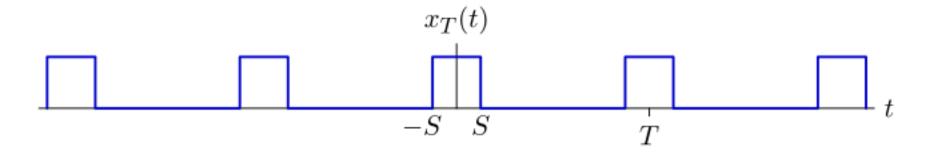
Represent $x_T(t)$ by its Fourier series.



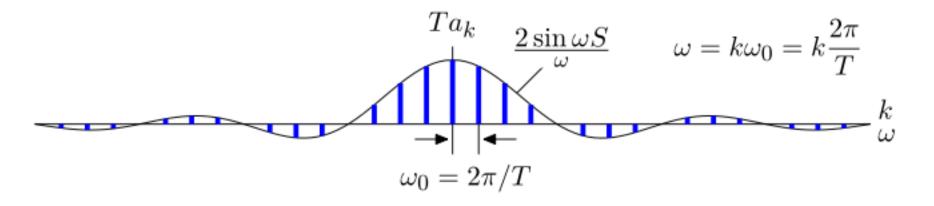
$$a_k = \frac{1}{T} \int_{-T/2}^{T/2} x_T(t) e^{-j\frac{2\pi}{T}kt} dt = \frac{1}{T} \int_{-S}^{S} e^{-j\frac{2\pi}{T}kt} dt = \frac{\sin\frac{2\pi kS}{T}}{\pi k} = \frac{2}{T} \frac{\sin\omega S}{\omega}$$



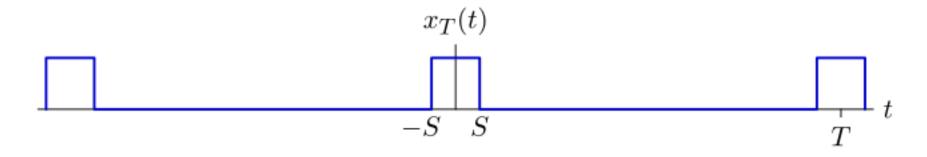
Doubling period doubles # of harmonics in given frequency interval.



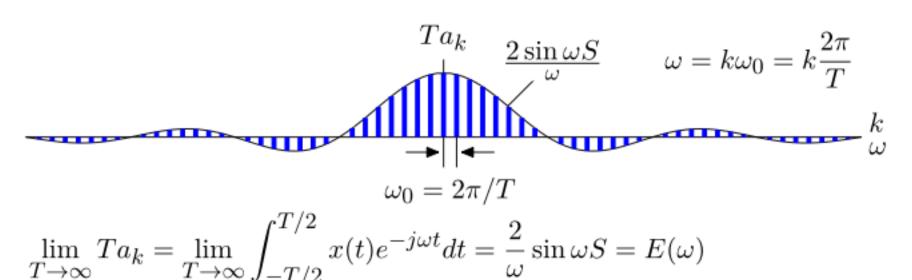
$$a_k = \frac{1}{T} \int_{-T/2}^{T/2} x_T(t) e^{-j\frac{2\pi}{T}kt} dt = \frac{1}{T} \int_{-S}^{S} e^{-j\frac{2\pi}{T}kt} dt = \frac{\sin\frac{2\pi kS}{T}}{\pi k} = \frac{2}{T} \frac{\sin\omega S}{\omega}$$



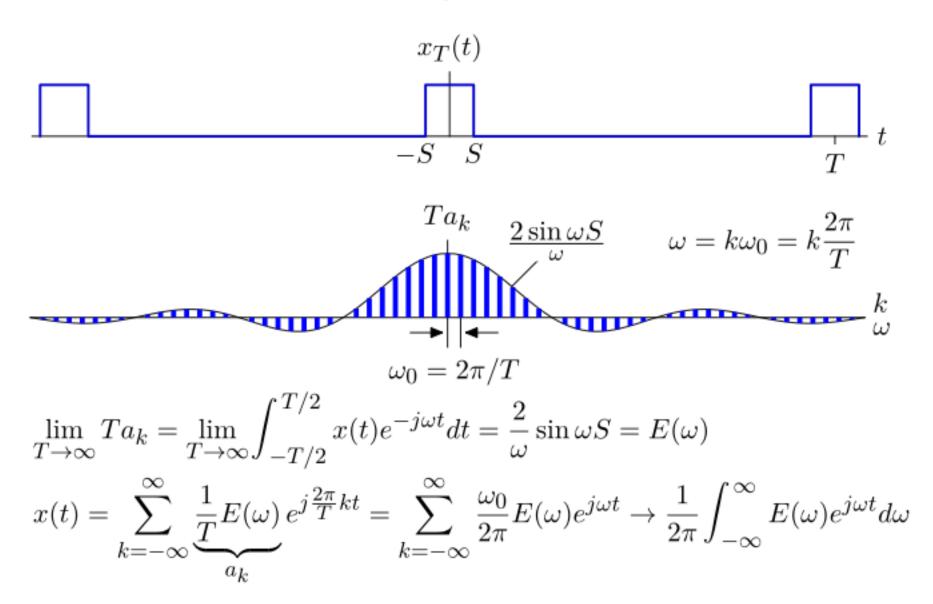
As $T \to \infty$, discrete harmonic amplitudes \to a continuum $E(\omega)$.



$$a_k = \frac{1}{T} \int_{-T/2}^{T/2} x_T(t) e^{-j\frac{2\pi}{T}kt} dt = \frac{1}{T} \int_{-S}^{S} e^{-j\frac{2\pi}{T}kt} dt = \frac{\sin\frac{2\pi kS}{T}}{\pi k} = \frac{2}{T} \frac{\sin\omega S}{\omega}$$



As $T \to \infty$, synthesis sum \to integral.



Replacing $E(\omega)$ by $X(j\omega)$ yields the Fourier transform relations.

$$E(\omega) = X(j\omega)$$

Fourier transform

$$X(j\omega) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t}dt$$
 ("analysis" equation)

$$x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\omega) e^{j\omega t} d\omega$$
 ("synthesis" equation)

Form is similar to that of Fourier series

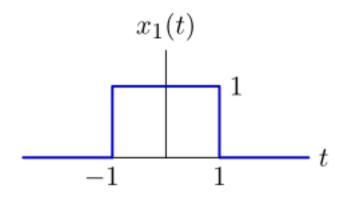
 \rightarrow provides alternate view of signal.

Properties of the Fourier Transform

Property	x(t)	$X(j\omega)$
Linearity	$ax_1(t) + bx_2(t)$	$aX_1(j\omega) + bX_2(j\omega)$
Time shift	$x(t-t_0)$	$e^{-j\omega t_0}X(j\omega)$
Time scale	x(at)	$\frac{1}{ a }X\left(\frac{j\omega}{a}\right)$
Differentiation	$\frac{dx(t)}{dt}$	$j\omega X(j\omega)$
Multiply by t	tx(t)	$-\frac{1}{j}\frac{d}{d\omega}X(j\omega)$
Convolution	$x_1(t) * x_2(t)$	$X_1(j\omega) \times X_2(j\omega)$

The Fourier transform is a function of real domain: frequency ω .

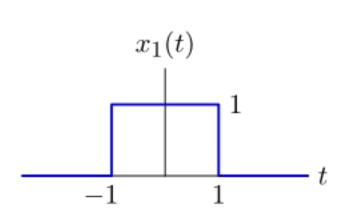
Time representation:

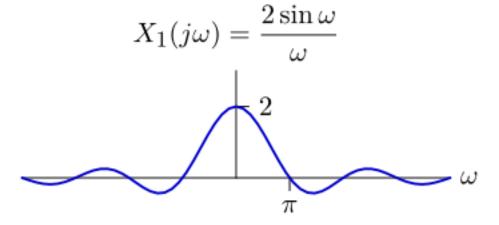


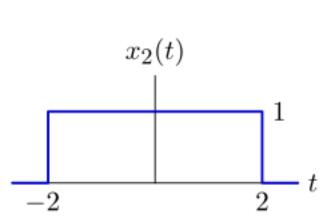
Frequency representation:

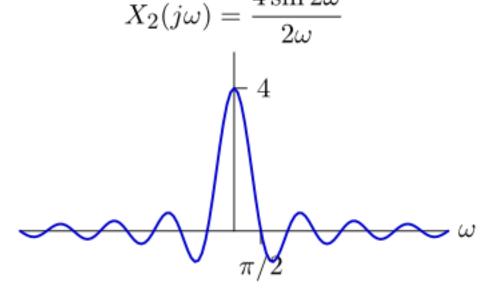
$$X_1(j\omega) = \frac{2\sin\omega}{\omega}$$

Stretching time compresses frequency.





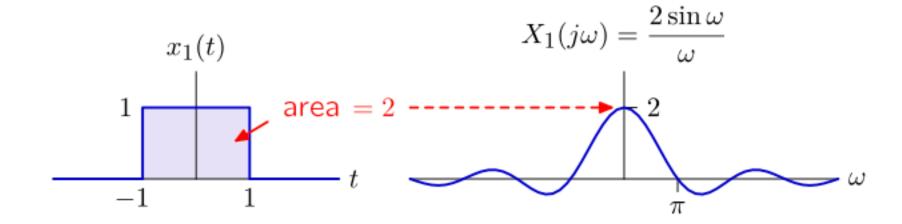




Moments

The value of $X(j\omega)$ at $\omega=0$ is the integral of x(t) over time t.

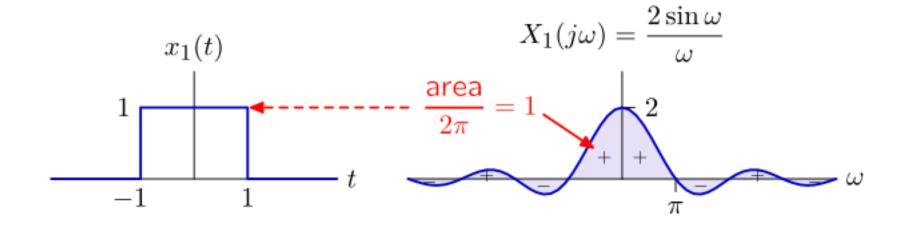
$$X(j\omega)|_{\omega=0} = \int_{-\infty}^{\infty} x(t)e^{-j\omega t}dt = \int_{-\infty}^{\infty} x(t)e^{j0t}dt = \int_{-\infty}^{\infty} x(t)\,dt$$



Moments

The value of x(0) is the integral of $X(j\omega)$ divided by 2π .

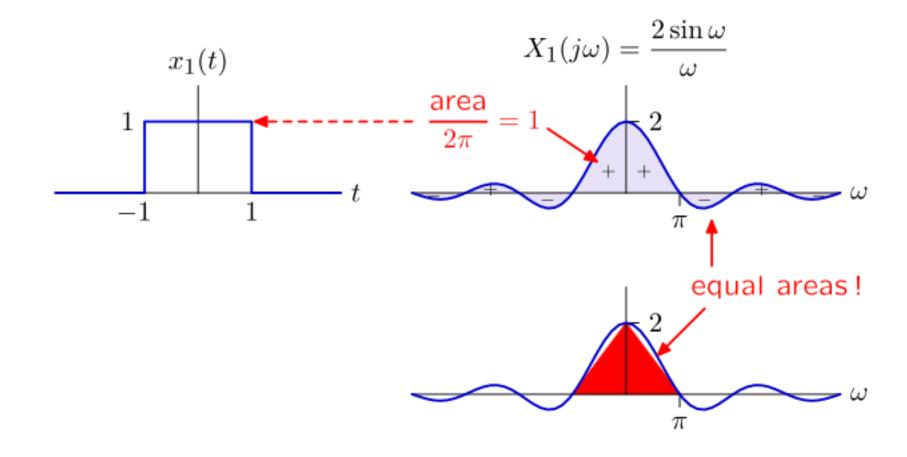
$$x(0) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\omega) \, e^{j\omega t} d\omega = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\omega) \, d\omega$$



Moments

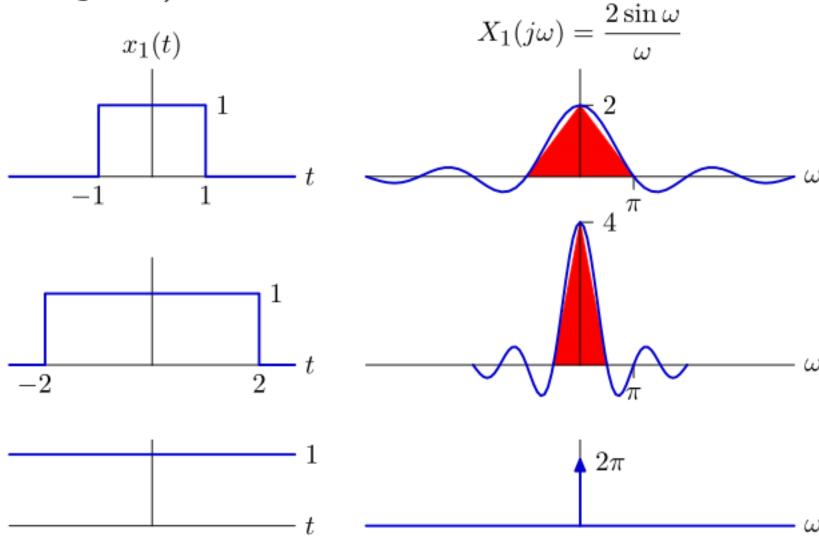
The value of x(0) is the integral of $X(j\omega)$ divided by 2π .

$$x(0) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\omega) \, e^{j\omega t} d\omega = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\omega) \, d\omega$$



Stretching to the Limit

Stretching time compresses frequency and increases amplitude (preserving area).



New way to think about an impulse!

One of the most useful features of the Fourier transform (and Fourier series) is the simple "inverse" Fourier transform.

$$X(j\omega) = \int_{-\infty}^{\infty} x(t)e^{-j\omega t}dt \qquad \qquad \text{(Fourier transform)}$$

$$x(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} X(j\omega)e^{j\omega t}d\omega \qquad \qquad \text{("inverse" Fourier transform)}$$

There are 4 Fourier Transforms!

• Recall Fourier series:

$$c_n = \frac{1}{T} \int_{-T/2}^{T/2} f(t)e^{-j\frac{2\pi n}{T}t}$$
$$f(t) \approx \sum_{n=-\infty}^{\infty} c_n e^{j\frac{2\pi n}{T}t}$$

- f(t) is periodic with period T!
- General Fourier Transform requires no periodicity:

$$F(\omega) = \int_{-\infty}^{\infty} f(t)e^{-j2\pi\omega t}dt$$

$$f(t) = \int_{-\infty}^{\infty} F(\omega) e^{j2\pi\omega t} dt$$

DFT — the most important one

 Discrete Fourier Transform (DFT) requires periodicity in both transform pairs

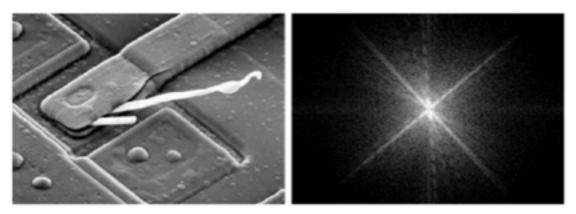
$$F_m = \sum_{n=0}^{M-1} f_n e^{-j2\pi mn/M}$$

$$f_n = \frac{1}{M} \sum_{m=0}^{M-1} F_m e^{j2\pi mn/M}$$

All Fourier Transforms

	Spatial Domain	Frequency Domain
FT	$f(t) = \int_{-\infty}^{\infty} F(\omega) e^{j2\pi\omega t} dt$ continuous	$F(\omega) = \int_{-\infty}^{\infty} f(t) e^{-j2\pi\omega t} dt$ continuous
FS — Fourier Series	$f(t) = \sum_{n=-\infty}^{\infty} c_n e^{j\frac{2\pi n}{T}t}$ continuous + periodic	$c_n = \frac{1}{T} \int_{-T/2}^{T/2} f(t) e^{-j\frac{2\pi n}{T}t} dt$ discrete
DFT — Discrete FT	$f_n = rac{1}{M} \sum_{m=0}^{M-1} F_m e^{j2\pi m n/M}$ discrete + periodic	$F_m = \sum_{n=0}^{M-1} f_n e^{-j2\pi mn/M}$ discrete + periodic
DTFT — Discrete Time FT	$f_n = \frac{1}{2\pi} \int_{-\pi}^{\pi} F(\omega) e^{j\omega n} d\omega$ discrete	$F(\omega) = \sum_{n=-\infty}^{\infty} f_n e^{-j\omega n} d\omega$ continuous + periodic

© Torsten Möller



a b

FIGURE 4.29 (a) SEM image of a damaged integrated circuit. (b) Fourier spectrum of (a). (Original image courtesy of Dr. J. M. Hudak, Brockhouse Institute for Materials Research, McMaster University, Hamilton, Ontario, Canada.)

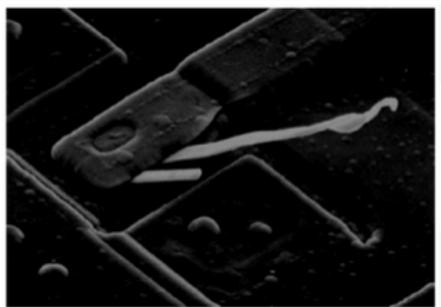
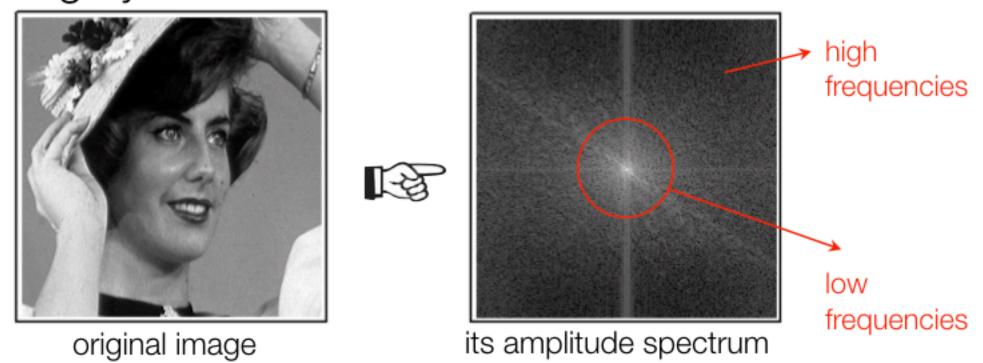


FIGURE 4.30 Result of filtering the image in Fig. 4.29(a) by setting to 0 the term F(M/2, N/2) in the Fourier transform.

Visualisation of the spectrum

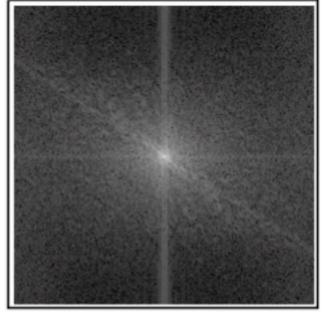
 Generally, we look at the amplitude of an image transform; hence we take a logarithmic scale to represent the values as gray-values.



<u>Interpretation</u>





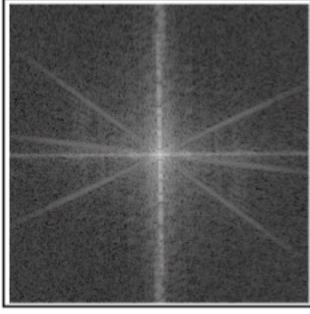


- The lines correspond to discontinuities, with perpendicular orientation.
- The horizontal and vertical lines come from the implicit periodic boundary conditions.
- For natural images, most of the information is concentrated in the low-frequency region.
- Low frequencies correspond to the slowly varying components, whereas high frequencies correspond to fast gray level changes (edges...)
- The diagonal line results from the discontinuity induced by the hat.

<u>Interpretation</u>



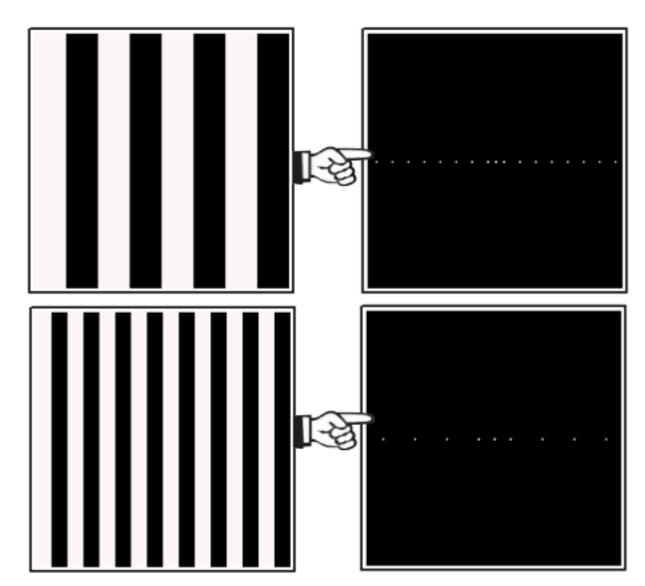




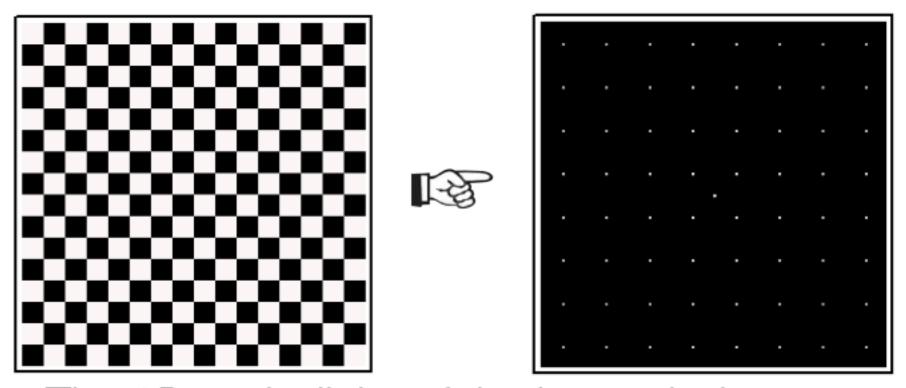
- The lines correspond to discontinuities, with perpendicular orientation.
- The horizontal and vertical lines come from the implicit periodic boundary conditions.
- For natural images, most of the information is concentrated in the low-frequency region.
- Low frequencies correspond to the slowly varying components, whereas high frequencies correspond to fast gray level changes (edges...)

Some Examples

- Since the image content is periodic, the spectrum is discrete.
- If the period of the signal increases, the distance between the frequencies decreases, and vice versa.



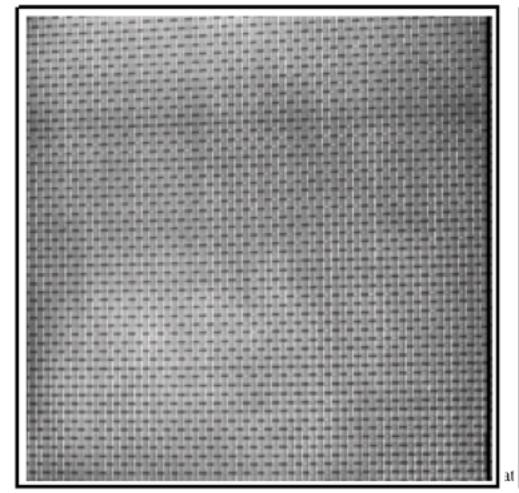
Some Examples

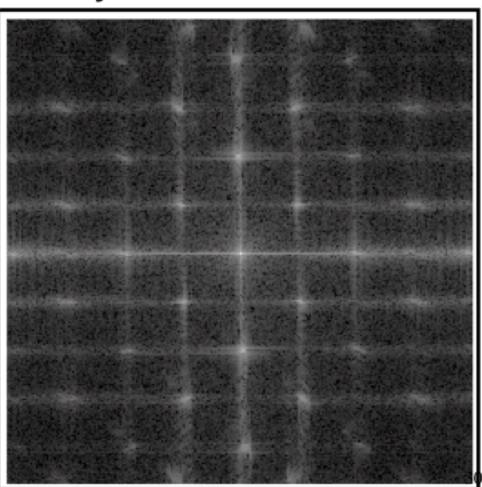


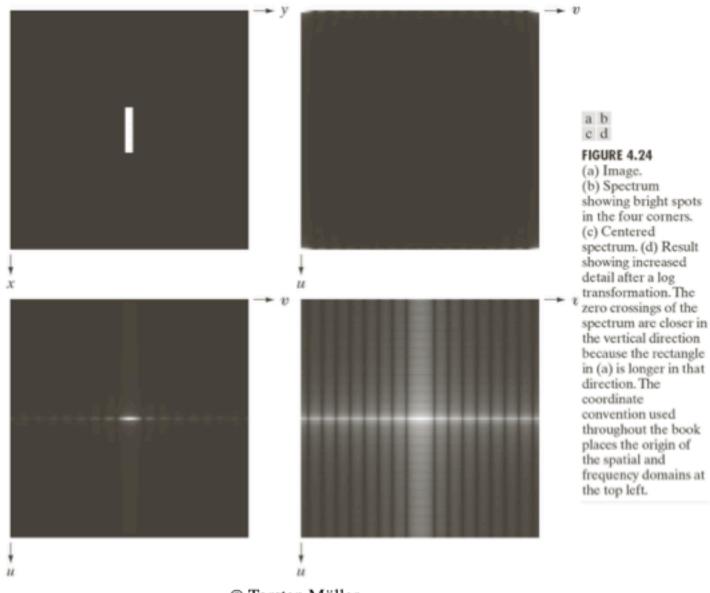
- The 2D periodicity of the image induces the 2D periodicity of the FT
- Since the image content is periodic, the spectrum is discrete.

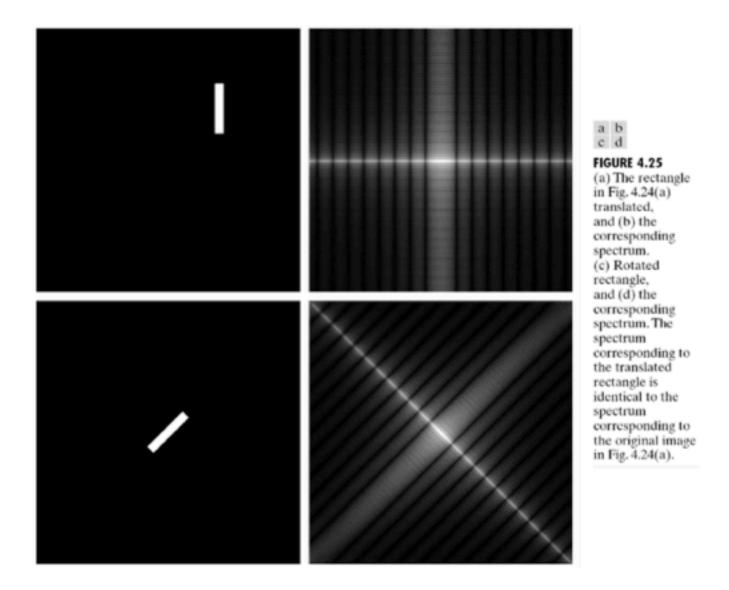
Texture = periodic pattern

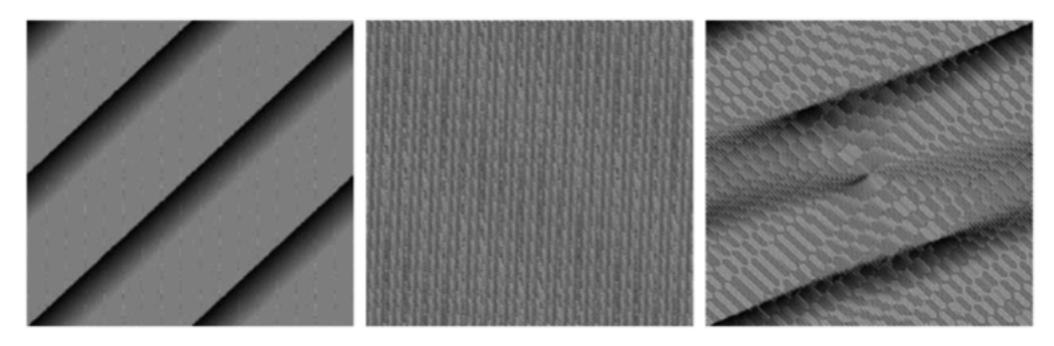
 The Fourier spectrum is well suited for describing the directionality of textures







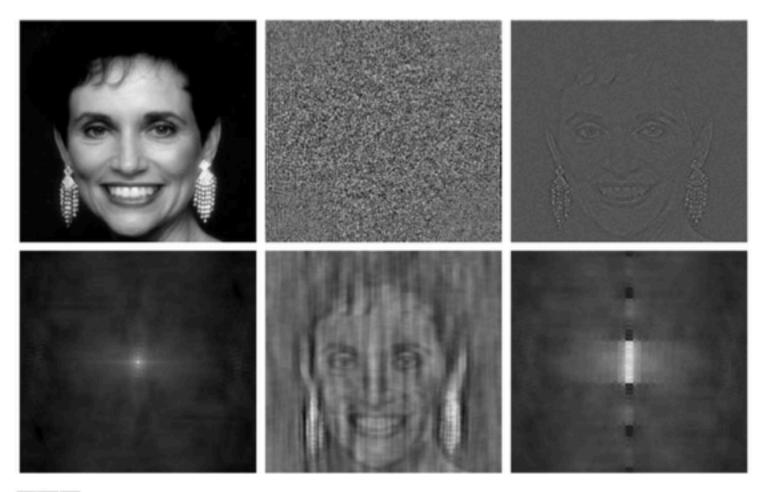




a b c

FIGURE 4.26 Phase angle array corresponding (a) to the image of the centered rectangle in Fig. 4.24(a), (b) to the translated image in Fig. 4.25(a), and (c) to the rotated image in Fig. 4.25(c).

Not very intuitive



a b c d e f

FIGURE 4.27 (a) Woman. (b) Phase angle. (c) Woman reconstructed using only the phase angle. (d) Woman reconstructed using only the spectrum. (e) Reconstruction using the phase angle corresponding to the woman and the spectrum corresponding to the rectangle in Fig. 4.24(a). (f) Reconstruction using the phase of the rectangle and the spectrum of the woman.

Learning Goals

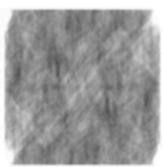
- What are the differences between the Fourier Series and the Fourier Transform?
- Why is the Fourier Transform so important?
- Low frequencies correspond to in an image?
- High frequencies correspond to in an image?

Recall

• What are the properties of $F(\omega)$ in the Fourier Transform?

 Which information contains the amplitude of a spectrum and which information contains the phase of a spectrum?



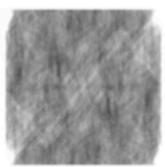




Recall

- Amplitude spectrum alone (middle image) and phase spectrum alone (rightmost image)
- with the phase spectrum alone, you can make sense of the nature of the image, even though the details are unclear.
- With the amplitude spectrum alone however, you cannot make any sense of the image, even though the dimensions(relative size of each component) are given.







Overview

- Fourier Series (1D)
 - motivation
 - properties
 - examples
- Sampling & Impulse Train
- Fourier Transform (1D)
 - properties
 - convolution theorem
 - sampling in the Fourier space

What happens to an impulse?

it is basically a constant!

$$c_n = \frac{1}{T} \int_{-T/2}^{T/2} \delta(t) e^{-j\frac{2\pi n}{T}t}$$

$$c_n = \frac{1}{T} e^0$$

$$c_n = \frac{1}{T}$$

What about a shifted impulse?

the shifts remain as frequencies

$$c_n = \frac{1}{T} \int_{-T/2}^{T/2} \delta(t - t_0) e^{-j\frac{2\pi n}{T}t}$$

$$c_n = \frac{1}{T} e^{-j\frac{2\pi n}{T}t_0}$$

What happens to an impulse train?

 Impulse train is periodic — apply Fourier series, will not do the math here, see book:

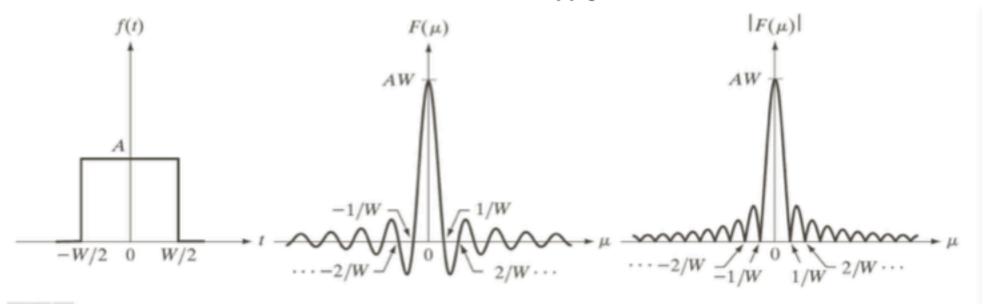
$$s_{\Delta T}(t) = \sum_{-\infty}^{\infty} \delta(t - n\Delta T)$$
$$S_{\Delta T}(\omega) = \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \delta(\omega - \frac{n}{\Delta T})$$

distance between impulses grows inversely

What happens to a box?

it is the well-known sinc function

$$\operatorname{sinc}(t) = \frac{\sin \pi t}{\pi t}$$



a b c

FIGURE 4.4 (a) A simple function; (b) its Fourier transform; and (c) the spectrum. All functions extend to infinity in both directions.

Overview

- Fourier Series (1D)
 - motivation
 - properties
 - examples
- Sampling & Impulse Train
- Fourier Transform (1D)
 - properties
 - convolution theorem
 - sampling in the Fourier space

LTI systems

$$f_s(t) = \sum_{n=-\infty}^{\infty} f[n]\delta(t - n\Delta T)$$



$$f_r(t) = \sum_{n=-\infty}^{\infty} f[n]\phi(t - n\Delta T)$$

LTI systems

- sine+cosine are eigenfunctions of an LTI system (a convolution)
- The eigenvalue popping out is the Fourier Transform of the LTI

$$e^{j\omega n\Delta T} \to \operatorname{Black\ Box}_{\operatorname{System}} \to e^{j\omega n\Delta T} \Phi(\omega)$$

$$\Phi(\omega) = \sum_{m=-\infty}^{\infty} e^{-mj\omega\Delta T} \phi(m\Delta T)$$

What is the Fourier Transform of a convolution?

doing this in the continuous domain:

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

$$F_r(\omega) = \int \left[\int f(\tau)h(t - \tau)d\tau\right] e^{-j2\pi\omega t}dt$$

$$= \int f(\tau) \left[\int h(t - \tau)e^{-j2\pi\omega t}dt\right]d\tau$$

$$= \int f(\tau) \left[H(\omega)e^{-j2\pi\omega \tau}\right]d\tau$$

$$= H(\omega)\int f(\tau)e^{-j2\pi\omega \tau}d\tau$$

$$= H(\omega)F(\omega) \qquad \text{© Torsten Möller}$$

What is the Fourier Transform of a convolution?

convolution == multiplication:

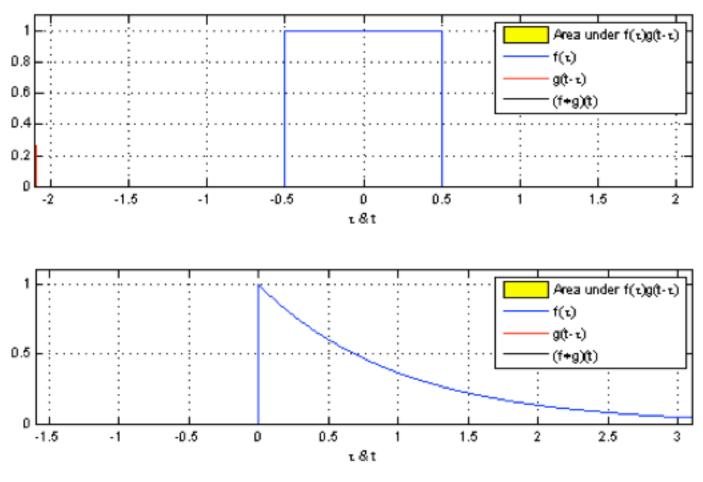
$$f * \phi(t) \iff F(\omega)H(\omega)$$

multiplication == convolution:

$$f(t)\phi(t) \iff F(\omega) * H(\omega)$$

Convolution — the movie

from wikipedia:



Learning Goals

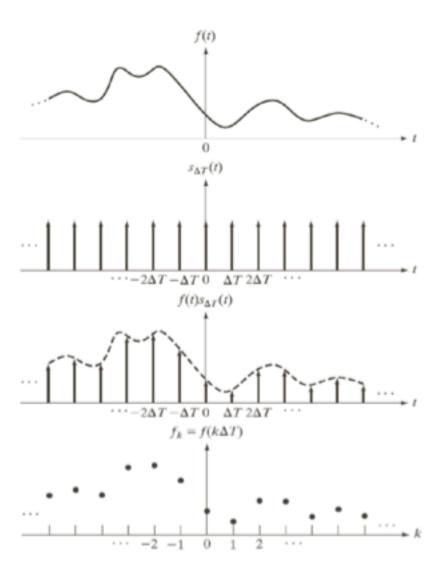
- Why is Convolution such an important concept?
- How does the Convolution work?
- What is an LTI System?
- How can I characterise an LTI System?

Overview

- Sampling & Impulse Train
- Fourier Transform (1D)
 - properties
 - convolution theorem
 - sampling in the Fourier space
- Fourier Transform (2D)

What was sampling again?

$$f(t) \times s_{\Delta T}(t) = \sum_{-\infty}^{\infty} f(n\Delta T)\delta(t - n\Delta T)$$
$$\sum_{-\infty}^{\infty} f[n]\delta(t - n\Delta T)$$



Sampling in the Fourier Domain

$$f(t)s_{\Delta T}(t) \iff F(\omega) * S_{\Delta T}(\omega)$$

$$F(\omega) * S_{\Delta T}(\omega) = F(\omega) * \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} \delta(\omega - \frac{n}{\Delta T})$$

$$= \frac{1}{\Delta T} \sum_{n=-\infty}^{\infty} F(\omega - \frac{n}{\Delta T})$$

Sampling allows the use of modern digital electronics to process, record, transmit, store, and retrieve CT signals.

audio: MP3, CD, cell phone

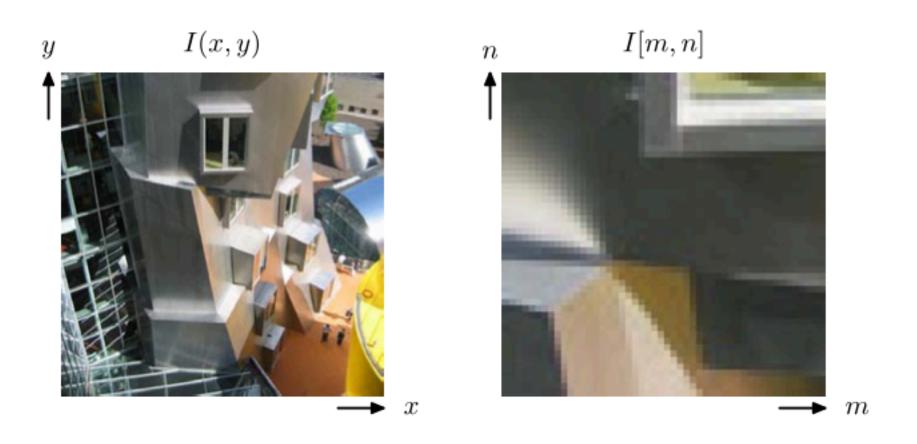
pictures: digital camera, printer

video: DVD

everything on the web

Sampling is pervasive.

Example: digital cameras record sampled images.

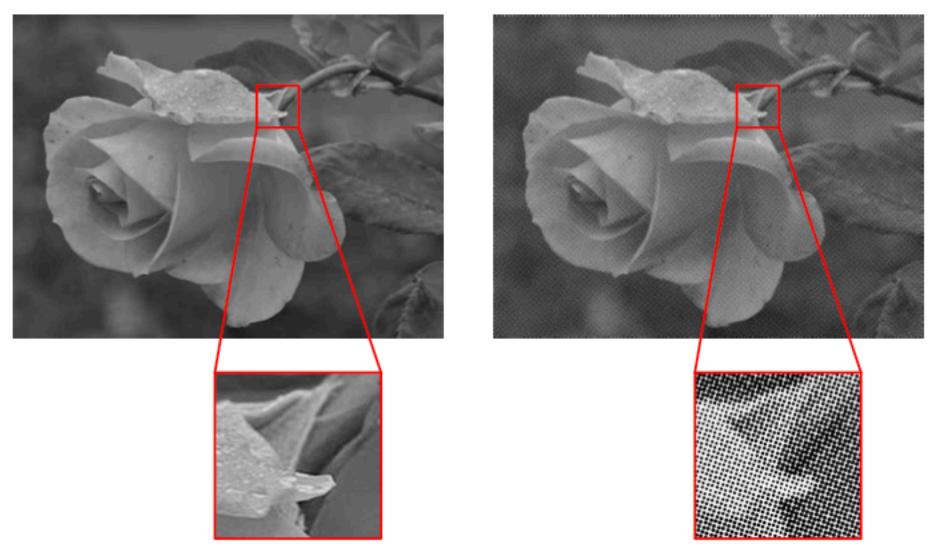


Photographs in newsprint are "half-tone" images. Each point is black or white and the average conveys brightness.



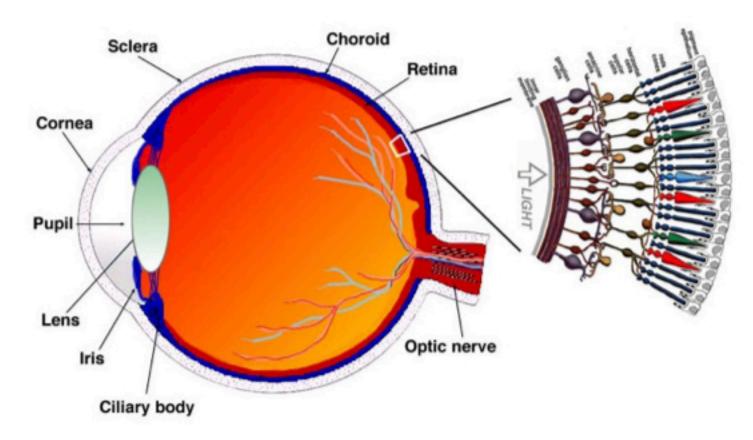


Zoom in to see the binary pattern.



© Prof. Dennis Freeman, MIT, 2011

Every image that we see is sampled by the retina, which contains \approx 100 million rods and 6 million cones (average spacing $\approx 3\mu$ m) which act as discrete sensors.



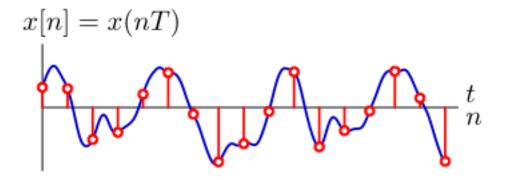
Courtesy of Helga Kolb, Eduardo Fernandez, and Ralph Nelson. Used with permission.

http://webvision.med.utah.edu/imageswv/sagschem.jpeg

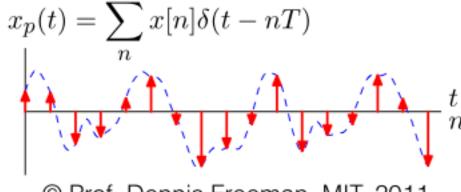
Sampling and Reconstruction

To determine the effect of sampling, compare the original signal x(t) to the signal $x_p(t)$ that is **reconstructed** from the samples x[n].

Uniform sampling (sampling interval T).



Impulse reconstruction.



Reconstruction

Impulse reconstruction maps samples x[n] (DT) to $x_p(t)$ (CT).

$$x_p(t) = \sum_{n = -\infty}^{\infty} x[n]\delta(t - nT)$$

$$= \sum_{n = -\infty}^{\infty} x(nT)\delta(t - nT)$$

$$= \sum_{n = -\infty}^{\infty} x(t)\delta(t - nT)$$

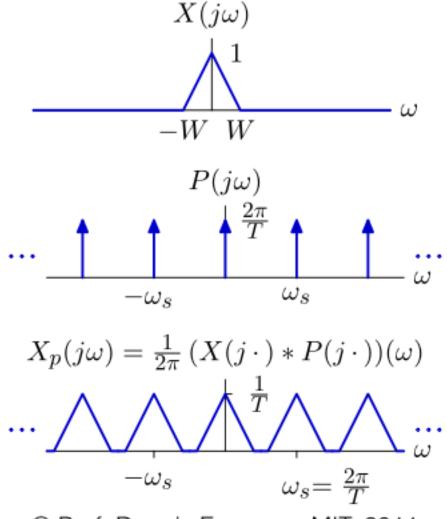
$$= x(t) \sum_{n = -\infty}^{\infty} \delta(t - nT)$$

$$\equiv x(t)$$

Resulting reconstruction $x_p(t)$ is equivalent to multiplying x(t) by impulse train.

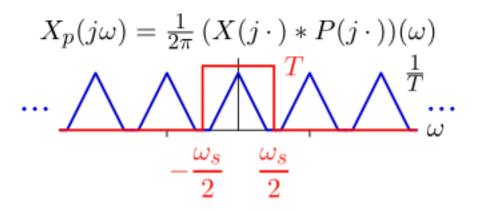
Multiplication by an impulse train in time is equivalent to convolution by an impulse train in frequency.

→ generates multiple copies of original frequency content.



© Prof. Dennis Freeman, MIT, 2011

The high frequency copies can be removed with a low-pass filter (also multiply by T to undo the amplitude scaling).



Impulse reconstruction followed by ideal low-pass filtering is called **bandlimited reconstruction**.

The Sampling Theorem

If signal is bandlimited \rightarrow sample without loosing information.

If x(t) is bandlimited so that

$$X(j\omega) = 0$$
 for $|\omega| > \omega_m$

then x(t) is uniquely determined by its samples x(nT) if

$$\omega_s = \frac{2\pi}{T} > 2\omega_m.$$

The minimum sampling frequency, $2\omega_m$, is called the "Nyquist rate."

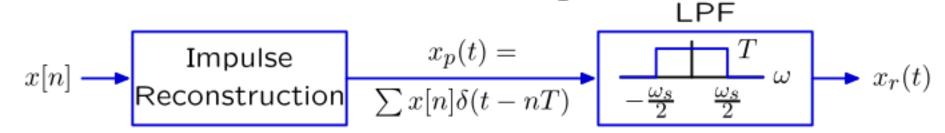
Summary

Three important ideas.

Sampling

$$x(t) \to x[n] = x(nT)$$

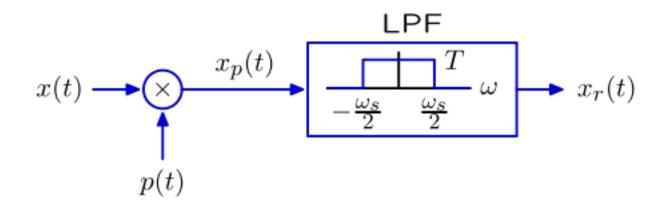
Bandlimited Reconstruction $\left(\omega_s = \frac{2\pi}{T}\right)$

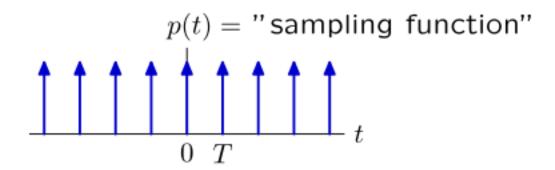


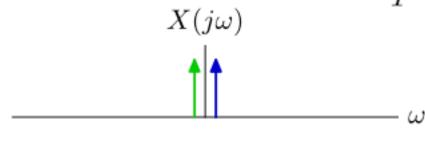
Sampling Theorem: If $X(j\omega) = 0 \ \forall \ |\omega| > \frac{\omega_s}{2}$ then $x_r(t) = x(t)$.

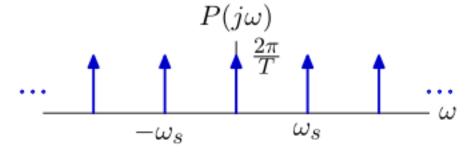
CT Model of Sampling and Reconstruction

Sampling followed by bandlimited reconstruction is equivalent to multiplying by an impulse train and then low-pass filtering.

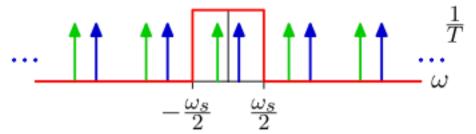


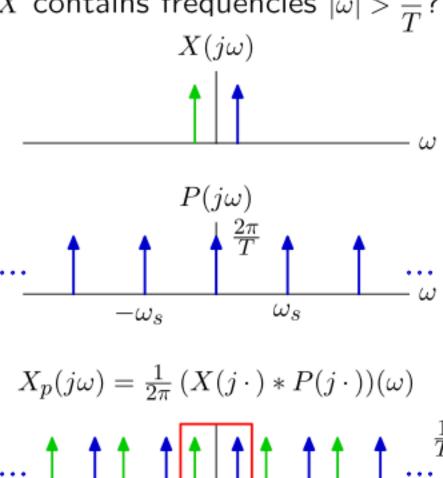


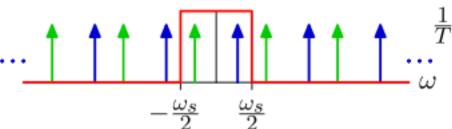


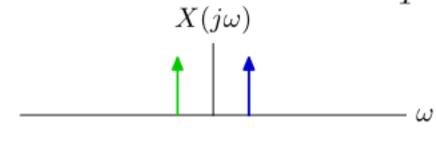


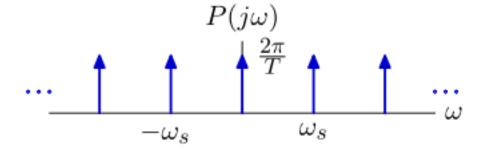
$$X_p(j\omega) = \frac{1}{2\pi} \left(X(j \cdot) * P(j \cdot) \right) (\omega)$$



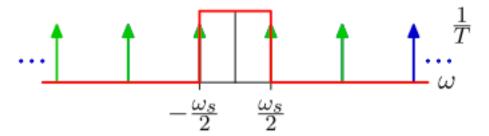


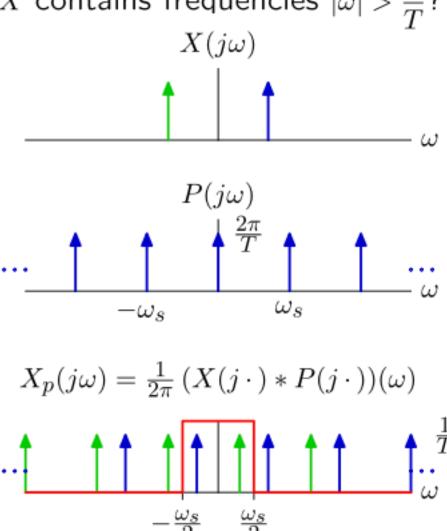


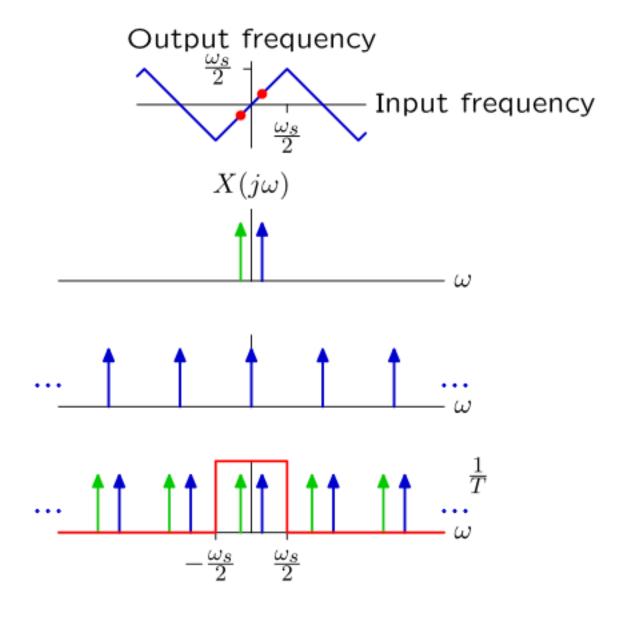




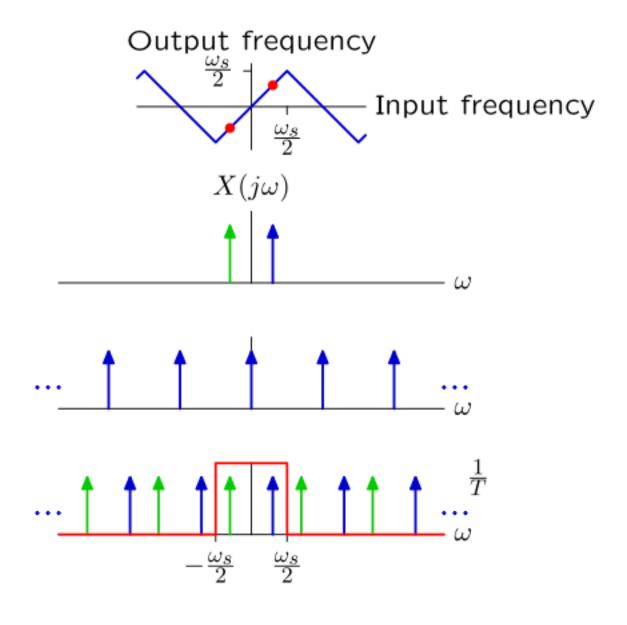
$$X_p(j\omega) = \frac{1}{2\pi} \left(X(j \cdot) * P(j \cdot) \right) (\omega)$$



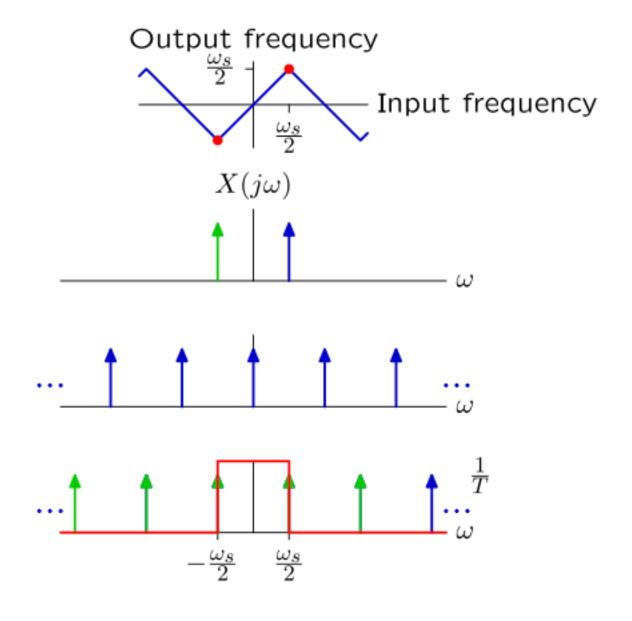


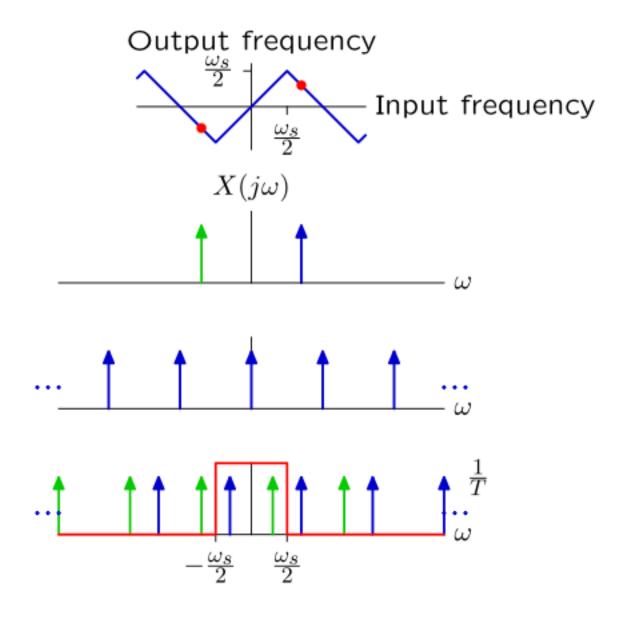


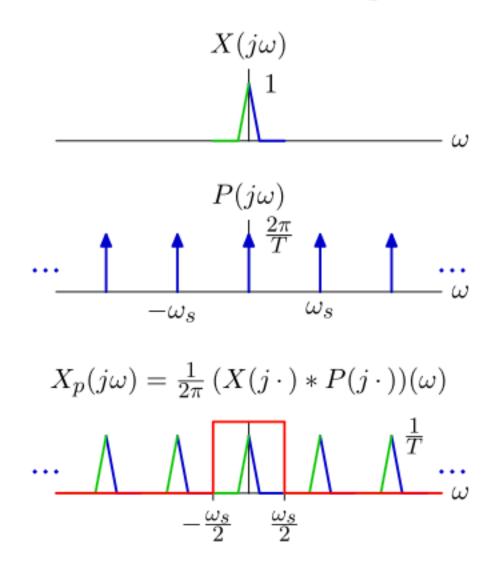
© Prof. Dennis Freeman, MIT, 2011

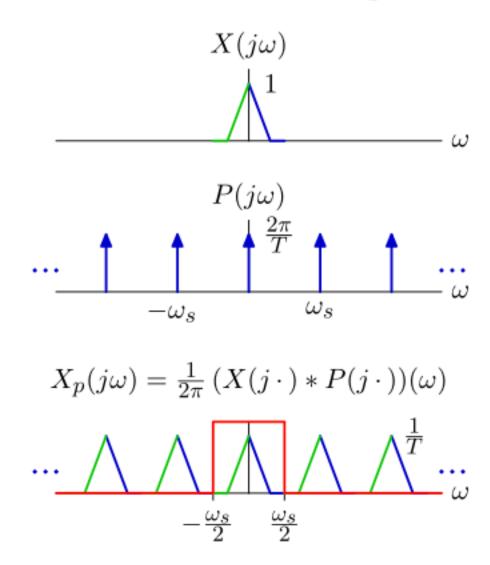


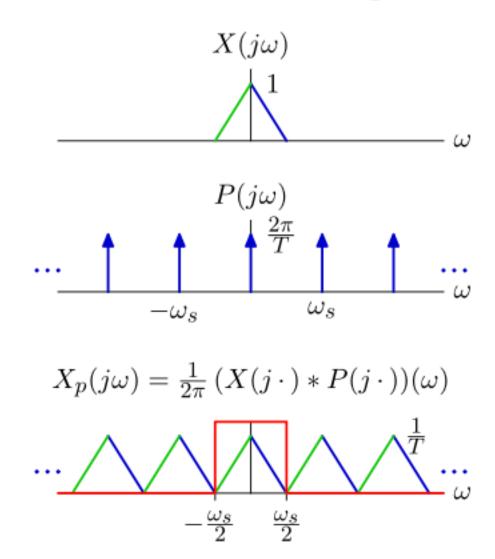
© Prof. Dennis Freeman, MIT, 2011

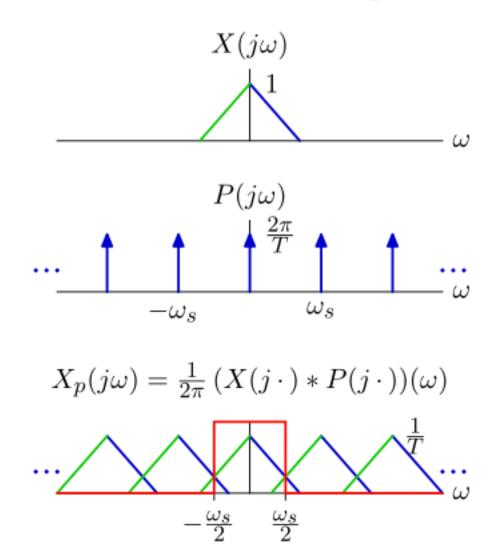


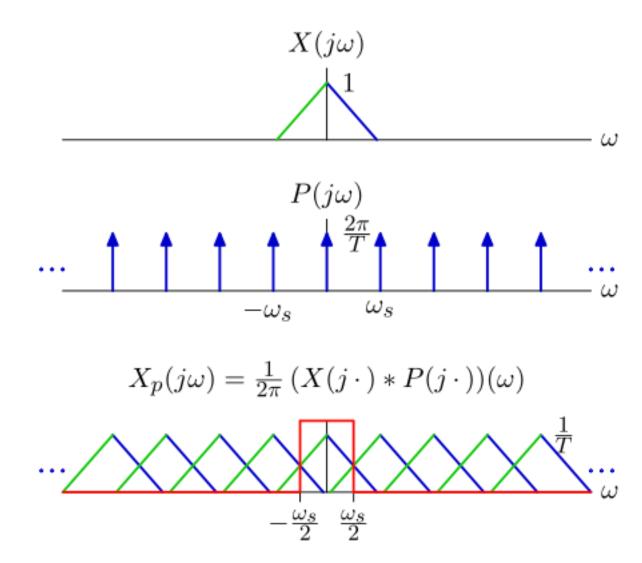






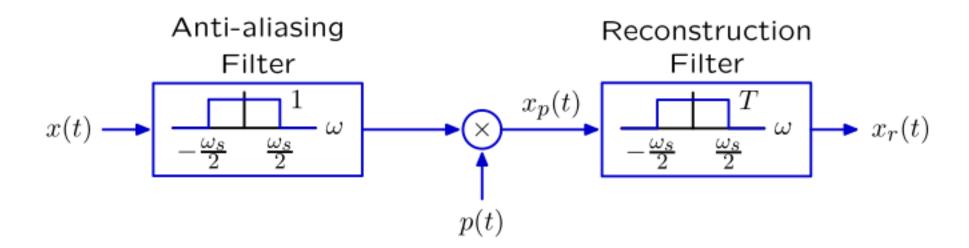


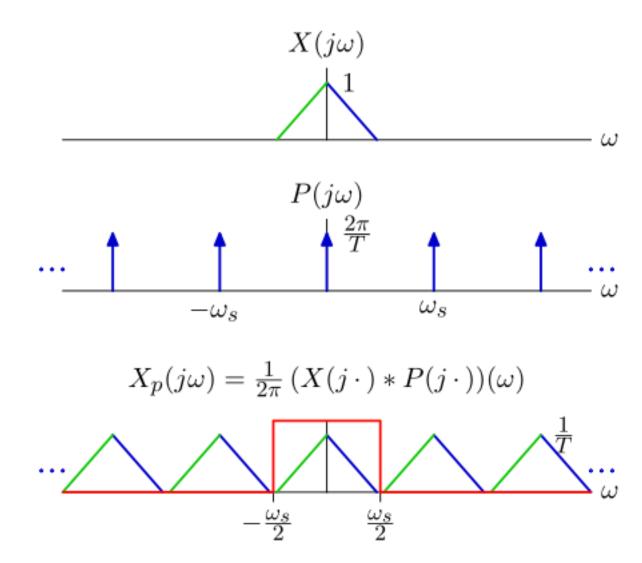


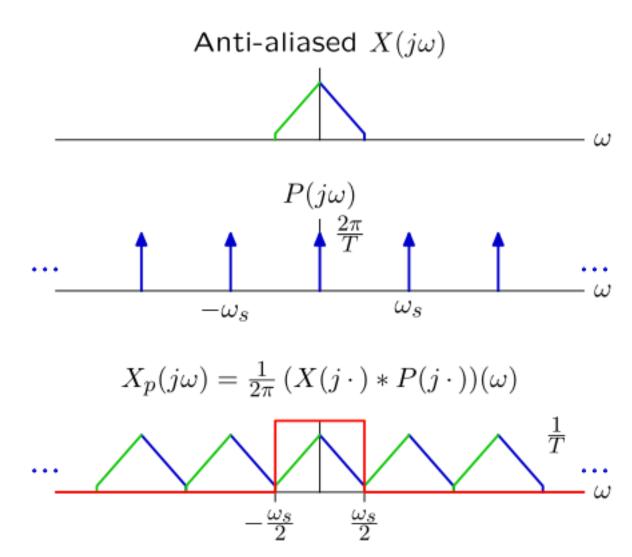


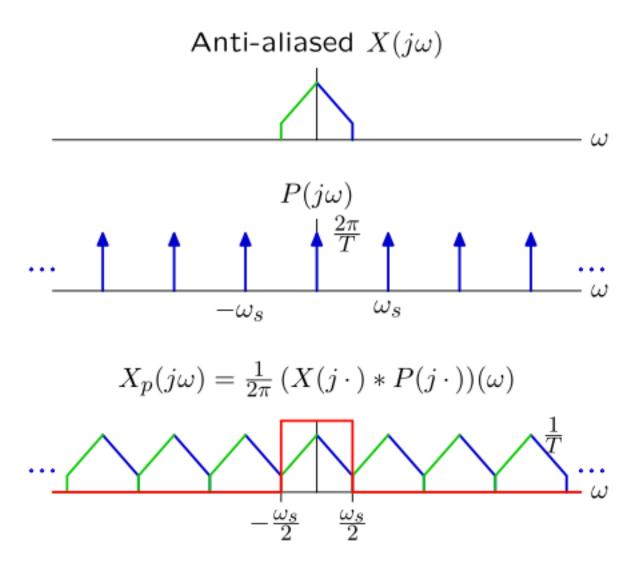
Anti-Aliasing Filter

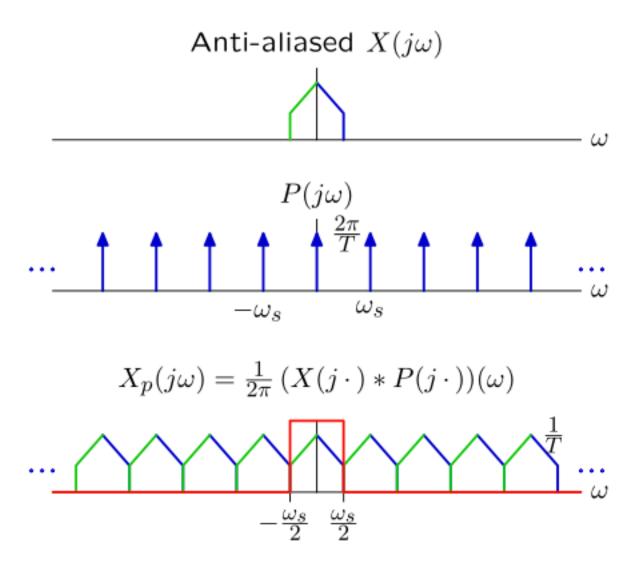
To avoid aliasing, remove frequency components that alias before sampling.













a b c

FIGURE 4.17 Illustration of aliasing on resampled images. (a) A digital image with negligible visual aliasing. (b) Result of resizing the image to 50% of its original size by pixel deletion. Aliasing is clearly visible. (c) Result of blurring the image in (a) with a 3 × 3 averaging filter prior to resizing. The image is slightly more blurred than (b), but aliasing is not longer objectionable. (Original image courtesy of the Signal Compression Laboratory, University of California, Santa Barbara.)

Learning Goals

- What are the properties of an LTI system?
- What is the Sampling Theorem?
- What is Aliasing and how can we avoid it?