

SHORT TIME FOURIER TRANSFORMS

Erwin M. Bakker

Overview

- Fourier Transforms

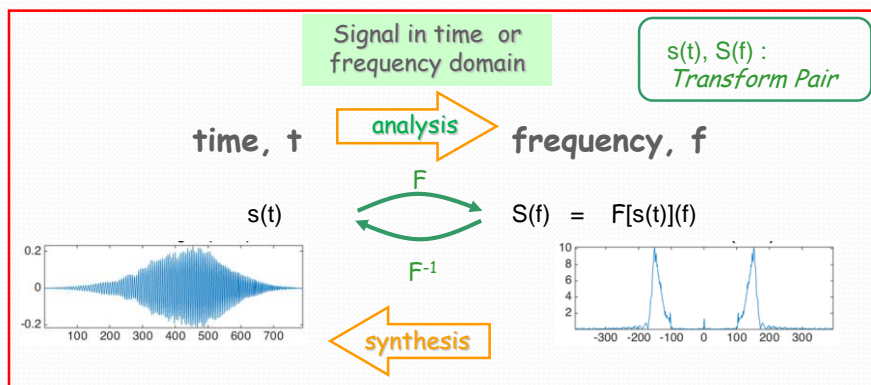
Material adapted from lectures by
Dr M.E. Angoletta at DISP2003,
a DSP course given by CERN and University of Lausanne (UNIL)

Fourier Transforms

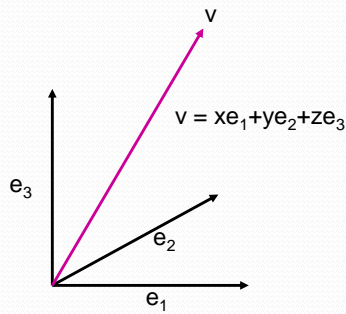
- Frequency analysis
- A tour of Fourier Transforms
- Continuous Fourier Series (FS)
- Discrete Fourier Series (DFS)

Frequency Analysis

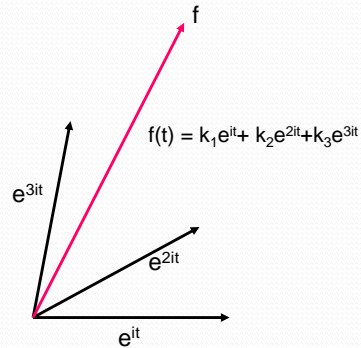
- **Fast & efficient insight on the signal's components.**
- **Powerful & complementary to time domain analysis techniques.**
- Simplifies the original problem - Filtering, solving Part.Diff.Eqns. (PDE),...
- Many transforms: **Fourier, Discrete Cosine, Laplace, z, Wavelet, etc.**



Bases of Vector Spaces

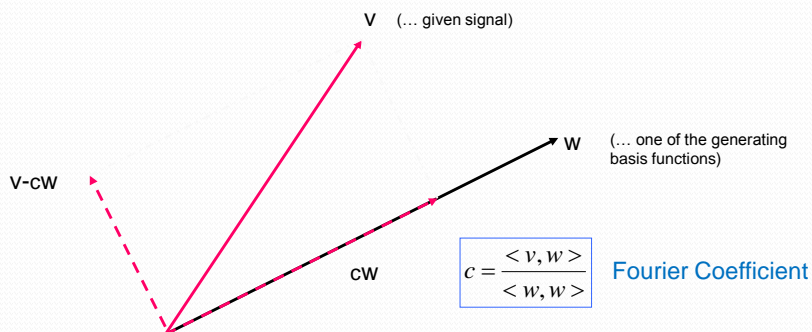


v is a linear combination of the basis **vectors** e_i ($i = 1, 2, 3$)



f is a linear combination of the basis **functions** e^{it}, e^{2it}, e^{3it}

Fourier Coefficients



Let $\langle \cdot, \cdot \rangle$ an in-product for our vector space V .
Then we calculate the Fourier coefficient c of v in V with respect to (basis) vector w by:

$$c = \frac{\langle v, w \rangle}{\langle w, w \rangle}$$

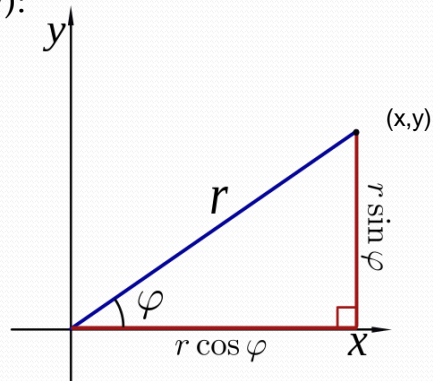
\Rightarrow cw is the component of v along the direction of w .

Polar Coordinates in \mathbb{R}^2

Relation between **Polar coordinates** (r, φ) and **Cartesian coordinates** (x, y) :

$$x = r \cos \varphi$$

$$y = r \sin \varphi$$



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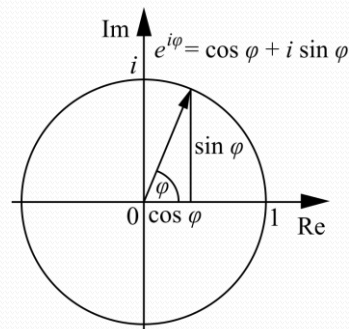
Complex Numbers

Define $e^{i\varphi} = \cos \varphi + i \sin \varphi$

Note: you can write any complex number

$z = a + bi$ as:

$$z = r e^{i\varphi}, \text{ with } r = |z|$$



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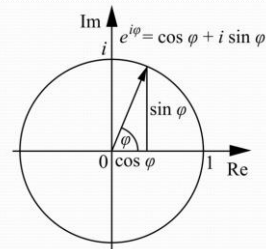
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Complex Numbers and Functions

Let $z = r e^{i\varphi}$, then $\bar{z} = r e^{-i\varphi}$ (alternative notation: z^*)

Let $z_1 = r_1 e^{-i\varphi_1}$, and $z_2 = r_2 e^{-i\varphi_2}$, then

$$z_1 z_2 = r_1 r_2 e^{-i(\varphi_1 + \varphi_2)}$$

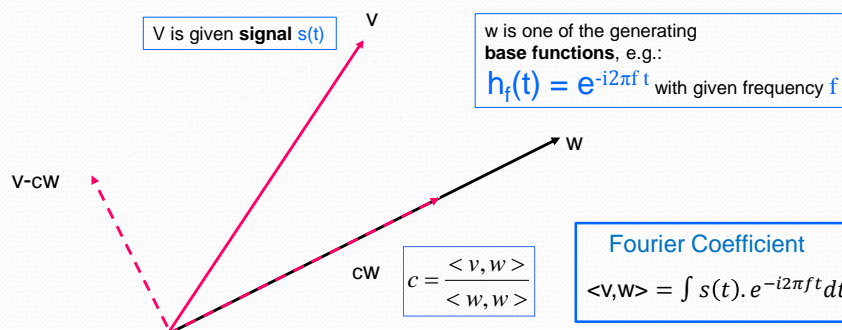


Let f a given frequency.

Let $h(t) = e^{i2\pi f t}$ then $h(t) = \cos 2\pi f t + i \sin 2\pi f t$, thus

$h(t)$ is a function that is 'repeating' over time with frequency f

Fourier Coefficients



Let $\langle \cdot, \cdot \rangle$ an in-product for our vector space V .

Then we calculate the Fourier coefficient c of v in V with respect to (basis) vector w by:

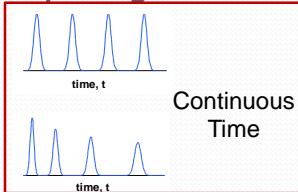
$$c = \frac{\langle v, w \rangle}{\langle w, w \rangle}$$

$\Rightarrow cw$ is the component of v along the direction of w .

Fourier Analysis – Different ‘Flavours’

Input Signal in Time Domain

Frequency spectrum



Continuous Time

Periodic (period T)
Aperiodic

FS

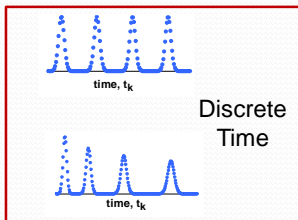
FT

Discrete

Continuous

$$c_k = \frac{1}{T} \cdot \int_0^T s(t) \cdot e^{-ik\omega t} dt$$

$$S(f) = \int_{-\infty}^{+\infty} s(t) \cdot e^{-i2\pi f t} dt$$



Discrete Time

Periodic (period T)

Aperiodic

D^{FS}**

DTFT

D^{FT}**

Discrete

Continuous

Discrete

$$\tilde{c}_k = \frac{1}{N} \sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{2\pi k n}{N}}$$

$$S(f) = \sum_{n=-\infty}^{+\infty} s[n] \cdot e^{-i2\pi f n}$$

$$\tilde{c}_k = \frac{1}{N} \sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{2\pi k n}{N}}$$

Note: $i = \sqrt{-1}$, $\omega = 2\pi/T$, $s[n] = s(t_n)$, $N = \text{No. of samples}$

** Calculated using FFT

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History Fourier Transform (1/2)

- **1669:** Newton: light spectra (*specter* = ghost) but no “frequency” concept (no waves).
- **18th century:** two important problems
 - celestial bodies orbits: Lagrange, Euler & Clairaut approximate observation data with linear combination of periodic functions; Clairaut, 1754(!) first DFT formula.
 - vibrating strings: Euler describes vibrating string motion by sinusoids (wave equation).
 - But consensus was: sum of sinusoids *only* represents smooth curves.
- **1807:** Fourier presents his work on heat conduction ⇒ Fourier analysis born.
 - Diffusion equation ⇔ series (infinite) of sines & cosines.
 - Strong criticism by peers blocks publication.
 - **Work published, 1822** (“*Theorie Analytique de la chaleur*”).

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History Fourier Transform (2/2)

➤ 19th / 20th century: two paths for Fourier analysis - Continuous & Discrete.

CONTINUOUS

- Fourier extends the analysis to arbitrary functions (Fourier Transform).
- Dirichlet, Poisson, Riemann, Lebesgue address Fourier Series convergence.
- Other FT variants born from varied needs (ex.: Short Time FT - speech analysis).

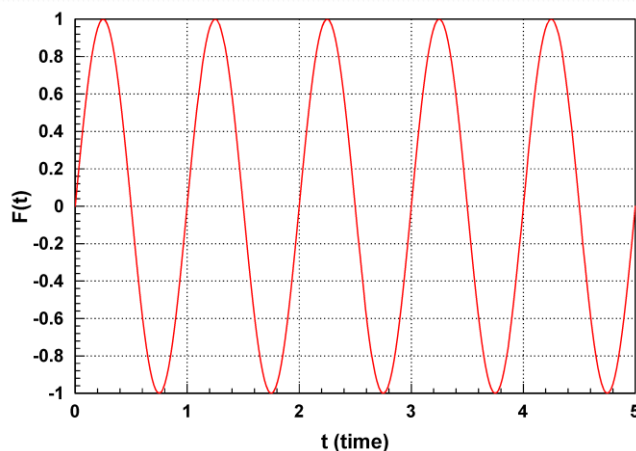
DISCRETE: Fast calculation methods (FFT)

- 1805 - Gauss, first usage of FFT (manuscript in Latin went unnoticed!!! Published 1866).
- 1965 - IBM's Cooley & Tukey "rediscover" FFT algorithm ("*An algorithm for the machine calculation of complex Fourier series*").
- Other DFT variants for different applications (ex.: Warped DFT - filter design & signal compression).
- FFT algorithm refined & modified for most computer platforms.
- *Fastest Fourier Transform in the West (FFTW)*

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Another Space, Another Base

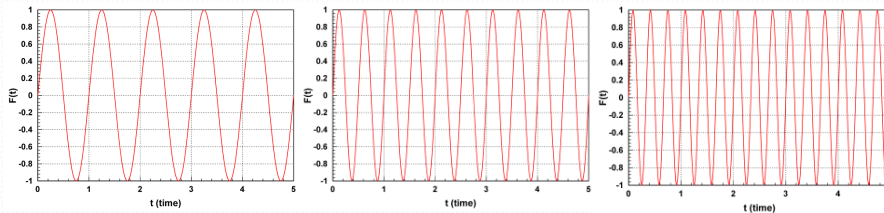


$$F(t) = \sin(2\pi.t)$$

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Another Space, Another Base



$$F(t) = \sin(2\pi.t)$$

$$F(t) = \sin(2\pi.2t)$$

$$F(t) = \sin(2\pi.3t)$$

$$F(t) = \cos(2\pi.t)$$

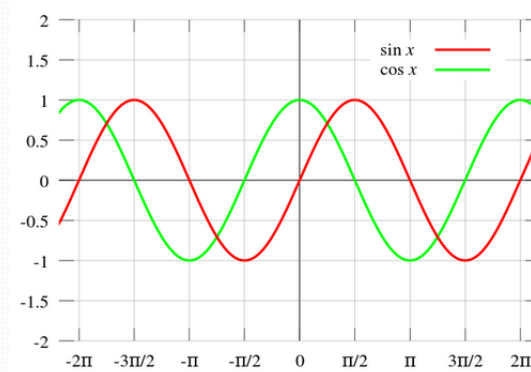
$$F(t) = \cos(2\pi.2t)$$

$$F(t) = \cos(2\pi.3t)$$

$\{ \cos(2\pi.kt), \sin(2\pi.kt) \}_k$ forms an orthogonal basis

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Sine Cosine Graphs

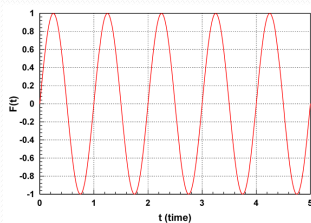


$$\sin(\varphi + \pi/2) = \cos(\varphi)$$

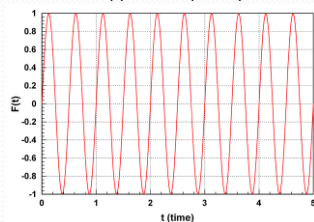
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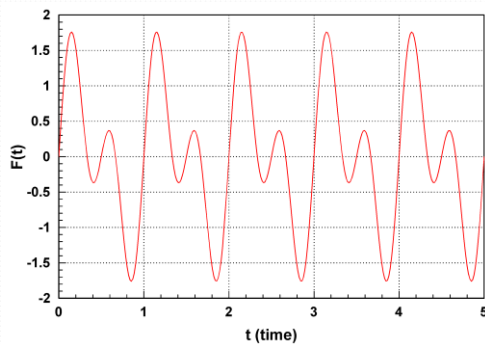
Linear Combination of Functions



$$F(t) = \sin(2\pi.t)$$



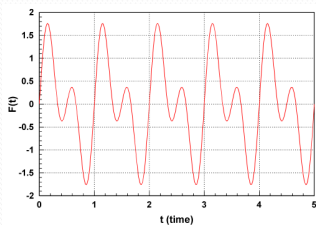
$$F(t) = \sin(2\pi.2t)$$



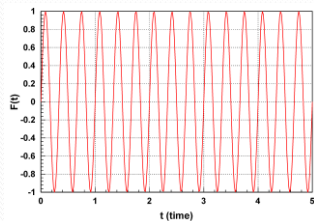
$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t)$$

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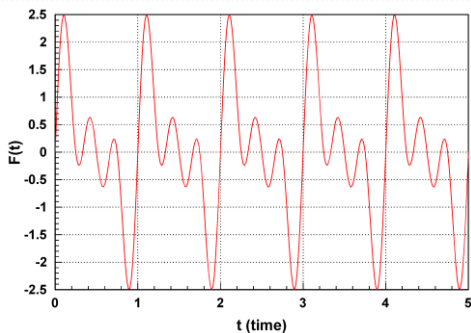
Linear Combination of Functions



$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t)$$



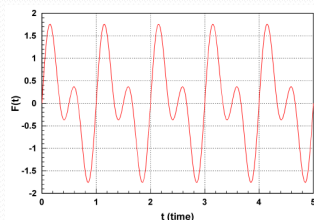
$$F(t) = \sin(2\pi.3t)$$



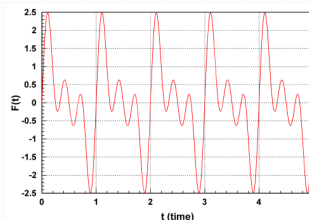
$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t) + \sin(2\pi.3t)$$

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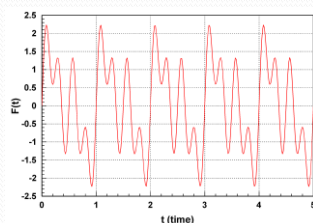
Linear Combination of Functions



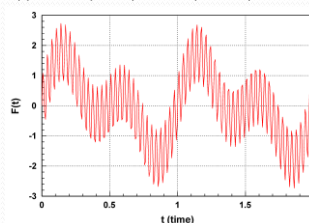
$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t)$$



$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t) + \sin(2\pi.3t)$$



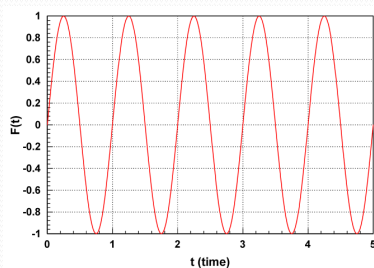
$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t) + \sin(2\pi.4t)$$



$$F(t) = \sin(2\pi.t) + \sin(2\pi.2t) + \sin(2\pi.30t)$$

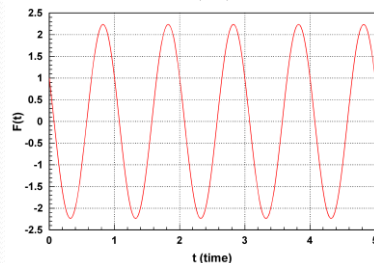
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Linear Combination of Functions



Phase Shift:

$$F(t) = \sin(2\pi.t)$$

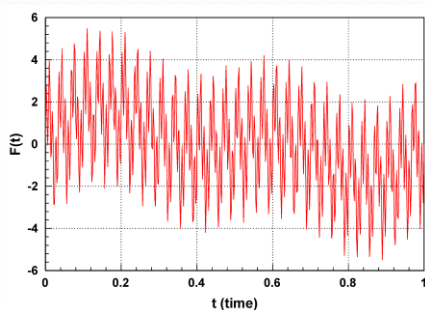


$$F(t) = \cos(2\pi.t) - 2.\sin(2\pi.t)$$

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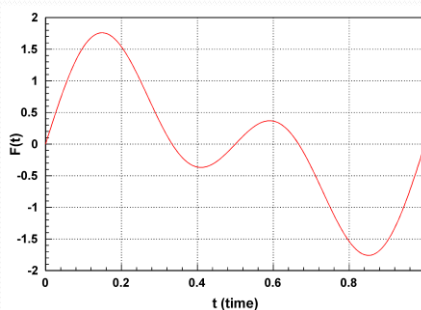
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Low Band Pass Filters



$$F(t) = \sin(2\pi \cdot t) + \sin(2\pi \cdot 2t) + \sin(2\pi \cdot 30t) + \sin(2\pi \cdot 120t)$$

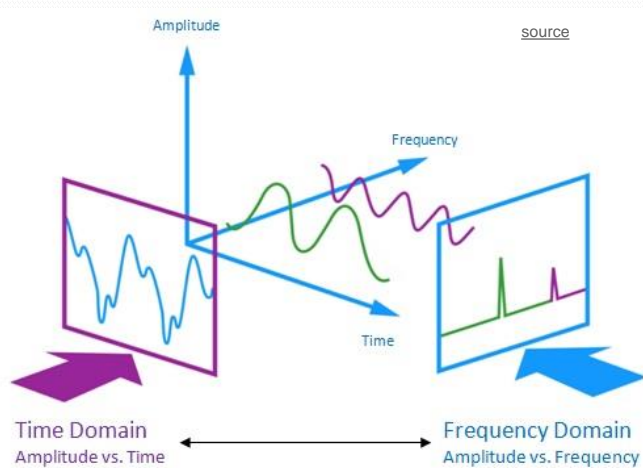
<- low freq.
<- high freq.



$$F(t) = \sin(2\pi \cdot t) + \sin(2\pi \cdot 2t)$$

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Fourier Series



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Fourier Series (FS)

* see next slide

A periodic function $s(t)$ satisfying Dirichlet's conditions * can be expressed as a Fourier series, with harmonically related sine/cosine terms.

synthesis

$$s(t) = a_0 + \sum_{k=1}^{+\infty} [a_k \cdot \cos(k\omega t) - b_k \cdot \sin(k\omega t)]$$

Inverse Fourier Transform

For all t but discontinuities

t : ~ time

a_0, a_k, b_k : Fourier coefficients.

k : ~ frequency, harmonic number

T : period, $\omega = 2\pi/T$

analysis

$$a_0 = \frac{1}{T} \cdot \int_0^T s(t) dt$$

Fourier Transform

(a_0 is signal average over a period,
i.e. Direct Current (DC) term & zero-frequency component.)

$$a_k = \frac{2}{T} \cdot \int_0^T s(t) \cdot \cos(k\omega t) dt$$

$$-b_k = \frac{2}{T} \cdot \int_0^T s(t) \cdot \sin(k\omega t) dt$$

Note: $\{\cos(k\omega t), \sin(k\omega t)\}_k$
form orthogonal base of
function space.

$$s(t) \leftrightarrow S(k) = (a_k, b_k)$$

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Fourier Series Convergence

Dirichlet conditions

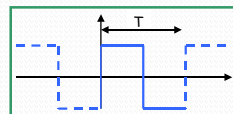
In any period:

(a) $s(t)$ piecewise-continuous;

(b) $s(t)$ piecewise-monotonic;

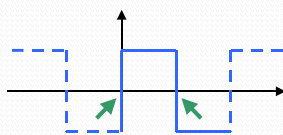
(c) $s(t)$ absolutely integrable, $\int_0^T |s(t)| dt < \infty$

Example:
square wave

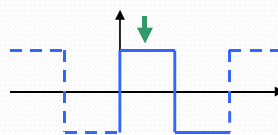


Rate of convergence

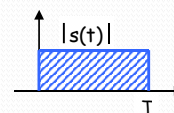
if $s(t)$ discontinuous then
 $|a_k| < M/k$ for large k ($M > 0$)



(a) ✓



(b) ✓



(c) ✓

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Fourier Series Analysis - 1

Fourier series of square wave $sw(t)$:

$$a_0 = \frac{1}{2\pi} \left\{ \int_0^{\pi} dt + \int_{\pi}^{2\pi} (-1) dt \right\} = 0 \quad (\text{zero average})$$

$$a_k = \frac{1}{\pi} \left\{ \int_0^{\pi} \cos kt \, dt - \int_{\pi}^{2\pi} \cos kt \, dt \right\} = 0$$

$$-b_k = \frac{1}{\pi} \left\{ \int_0^{\pi} \sin kt \, dt - \int_{\pi}^{2\pi} \sin kt \, dt \right\} = \dots = \frac{2}{k \cdot \pi} \cdot (1 - \cos k\pi) =$$

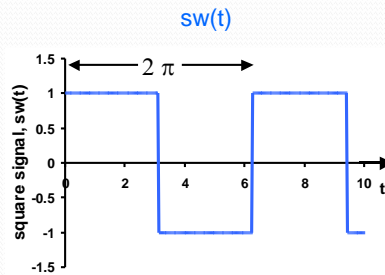
$$= \begin{cases} \frac{4}{k \cdot \pi}, & k \text{ odd} \\ 0, & k \text{ even} \end{cases}$$

$$sw(t) = \frac{4}{\pi} \cdot \sin t + \frac{4}{3 \cdot \pi} \cdot \sin 3 \cdot t + \frac{4}{5 \cdot \pi} \cdot \sin 5 \cdot t + \dots$$

$$a_0 = \frac{1}{T} \int_0^T s(t) dt$$

$$a_k = \frac{2}{T} \int_0^T s(t) \cdot \cos(k\omega t) dt$$

$$-b_k = \frac{2}{T} \int_0^T s(t) \cdot \sin(k\omega t) dt$$



$$T = 2\pi \Rightarrow \omega = 1$$

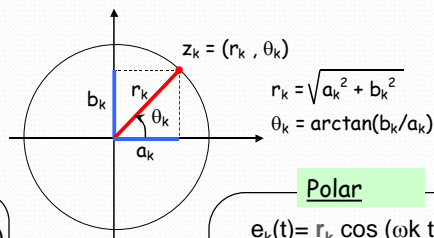
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Fourier Series Analysis - 2

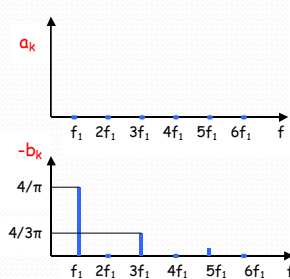
Fourier spectrum representations (in k)

$$s(t) = \sum_{k=0}^{\infty} v_k(t)$$



Rectangular

$$e_k(t) = a_k \cos(\omega_k t) - b_k \sin(\omega_k t)$$



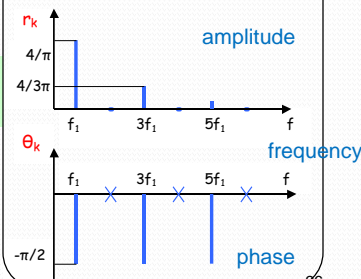
r_k = amplitude,
 θ_k = phase

$$f_k = k \omega / 2\pi$$

Fourier spectrum of square-wave.

Polar

$$e_k(t) = r_k \cos(\omega_k t + \theta_k)$$



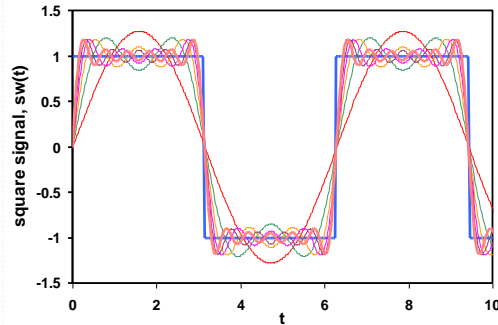
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Fourier Series Synthesis

Square wave reconstruction
from spectral terms

$$sw(t) = \sum_{k=1}^{\infty} \left[\frac{4}{k\pi} \sin(kt) \right]$$



Convergence may be slow ($\sim 1/k$) - ideally need infinite terms.

Practically, series truncated when remainder below computer tolerance
(\Rightarrow error). **BUT**... Gibbs' Phenomenon.

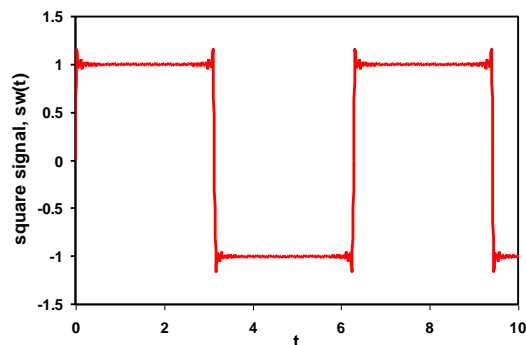
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Gibbs Phenomenon

Overshoot exist at each
discontinuity

$$sw_{79}(t) = \sum_{k=1}^{79} \left[-b_k \cdot \sin(kt) \right]$$



- First observed by Michelson, 1898. Explained by Gibbs.
- Max overshoot pk-to-pk = 8.95% of discontinuity magnitude.
- FS converges to $(-1+1)/2 = 0$ at discontinuities, *in this case*.

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Fourier Series Time Shifting

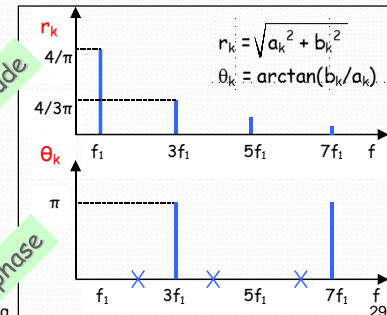
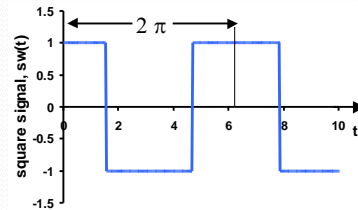
FS of even function:
 $\pi/2$ -advanced square-wave

$a_0 = 0$ (zero average)

$$a_k = \begin{cases} \frac{4}{k \cdot \pi} & , k \text{ odd}, k = 1, 5, 9, \dots \\ -\frac{4}{k \cdot \pi} & , k \text{ odd}, k = 3, 7, 11, \dots \\ 0 & , k \text{ even}. \end{cases}$$

$-b_k = 0$ (even function: $s(-x) = s(x)$)

Note: amplitudes unchanged **BUT**
 phases advance by $k \cdot \pi/2$.



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Complex Fourier Series

Euler's notation:

$e^{j\omega t} = (e^{j\omega t})^* = \cos(\omega t) - i \cdot \sin(\omega t)$

"phasor"

$\cos(\omega t) = \frac{e^{j\omega t} + e^{-j\omega t}}{2}$

$\sin(\omega t) = \frac{e^{j\omega t} - e^{-j\omega t}}{2 \cdot i}$

analysis

$$c_k = \frac{1}{T} \cdot \int_0^T s(t) \cdot e^{-ik\omega t} dt$$

synthesis

$$s(t) = \sum_{k=-\infty}^{\infty} c_k \cdot e^{ik\omega t}$$

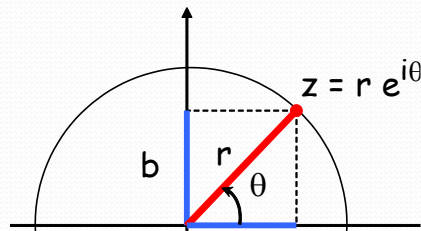
Complex form of FS (Laplace 1782). Harmonics c_k separated by $\Delta f = 1/T$ on frequency plot.

Note: $c_{-k} = (c_k)^*$

Link to FS real coeffs.

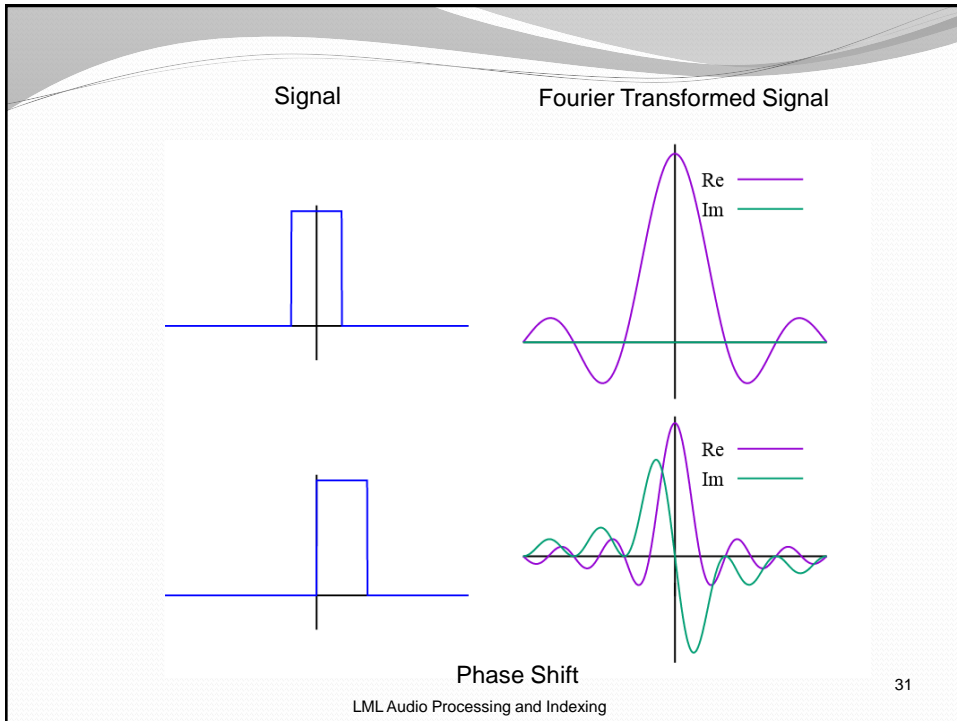
$c_0 = a_0$

$$c_k = \frac{1}{2} \cdot (a_k + i \cdot b_k) = \frac{1}{2} \cdot (a_{-k} - i \cdot b_{-k})$$



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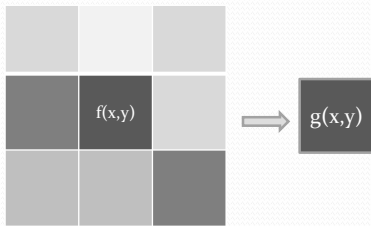
Fourier Series Properties

	Time (t)	Frequency (f)
Homogeneity	$a \cdot s(t)$	$a \cdot S(f)$
Additivity	$s(t) + u(t)$	$S(f) + U(f)$
Linearity	$a \cdot s(t) + b \cdot u(t)$	$a \cdot S(f) + b \cdot U(f)$
Time reversal	$s(-t)$	$S(-f)$
Multiplication	$s(t) \cdot u(t)$	$\frac{1}{T} \cdot \int_0^T S(f - t) \cdot U(t) dt$
Convolution	$\sum_{m=-\infty}^{\infty} s(m) u(t - m)$	$S(f) \cdot U(f)$
Time shifting	$s(t - \bar{t})$	$e^{-i \frac{2\pi f \cdot \bar{t}}{T}} \cdot S(f)$
Frequency shifting	$e^{+i \frac{2\pi m t}{T}} \cdot s(t)$	$S(f - m)$

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Image Processing: Convolutional Filters and Kernels



$$g(x,y) = \omega * f(x,y) = \sum_{dx=-a}^a \sum_{dy=-b}^b \omega(dx,dy) f(x+dx, y+dy)$$

Operation	Kernel ω	Image result $g(x,y)$
Identity	$\begin{bmatrix} 0 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & 0 & 0 \end{bmatrix}$	
Edge detection	$\begin{bmatrix} 1 & 0 & -1 \\ 0 & 0 & 0 \\ -1 & 0 & 1 \end{bmatrix}$	
	$\begin{bmatrix} 0 & -1 & 0 \\ -1 & 4 & -1 \\ 0 & -1 & 0 \end{bmatrix}$	
	$\begin{bmatrix} -1 & -1 & -1 \\ -1 & 8 & -1 \\ -1 & -1 & -1 \end{bmatrix}$	
Sharpen	$\begin{bmatrix} 0 & -1 & 0 \\ -1 & 5 & -1 \\ 0 & -1 & 0 \end{bmatrix}$	

[https://en.wikipedia.org/wiki/Kernel_\(image_processing\)](https://en.wikipedia.org/wiki/Kernel_(image_processing))

Discrete Fourier Series (DFS)

Band-limited signal $s[n]$, period = N .

DFS generate periodic c_k with same signal period

DFS defined as:

FT: analysis

$$\tilde{c}_k = \frac{1}{N} \sum_{n=0}^{N-1} s[n] \cdot e^{-i \frac{2\pi k n}{N}}$$

Note: $\tilde{c}_{k+N} = \tilde{c}_k \Leftrightarrow$ same period N
i.e. time periodicity propagates to frequencies!

IFT: synthesis

$$s[n] = \sum_{k=0}^{N-1} \tilde{c}_k \cdot e^{i \frac{2\pi k n}{N}}$$

Synthesis: finite sum \Leftarrow band-limited $s[n]$

Orthogonality in DFS:

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

$\delta_{ij} = \begin{cases} 0 & \text{if } i \neq j, \\ 1 & \text{if } i = j. \end{cases}$
 Kronecker's delta

N consecutive samples of $s[n]$ completely describe s in time or frequency domains.

Discrete Fourier Series Analysis

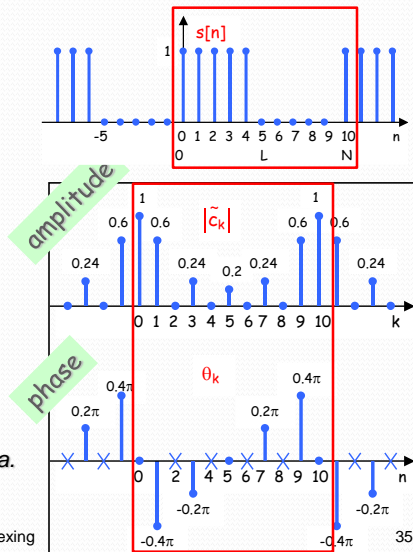
DFS of periodic discrete
1-Volt square-wave

$s[n]$: period N , duty factor L/N

$$\tilde{c}_k = \begin{cases} \frac{L}{N}, & k = 0, \pm N, \pm 2N, \dots \\ \frac{e^{-j\frac{\pi k(L-1)}{N}} \sin\left(\frac{\pi kL}{N}\right)}{N \sin\left(\frac{\pi k}{N}\right)}, & \text{otherwise} \end{cases}$$

Discrete signals \Rightarrow periodic frequency spectra.
Compare to continuous rectangular function
(slide # 20)

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Fourier Transforms

Let $s(\cdot)$ a signal in the **time domain**: $s(t)$ values as a function of **time** t ($-\infty < t < \infty$)

The same signal can be described as amplitudes and phases (complex values)

$S(\cdot)$ in the **frequency domain**: $S(f)$ values as a function of **frequency** f ($-\infty < f < \infty$)

One can transform the representation $s(t)$ in the **time domain** to the representation $S(f)$ in the **frequency domain** by using the Fourier Transform equation:

$$S(f) = \int_{-\infty}^{\infty} s(t) \cdot e^{-2\pi i f t} dt$$

And back, using the inverse FT-equation:

$$s(t) = \int_{-\infty}^{\infty} S(f) \cdot e^{2\pi i f t} df$$

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Some Discrete Fourier Series Properties

	Time (n)	Frequency (k)
Homogeneity	$a \cdot s[n]$	$a \cdot S(k)$
Additivity	$s[n] + u[n]$	$S(k) + U(k)$
Linearity	$a \cdot s[n] + b \cdot u[n]$	$a \cdot S(k) + b \cdot U(k)$
Multiplication	$s[n] \cdot u[n]$	$\frac{1}{N} \cdot \sum_{h=0}^{N-1} S(h)U(k-h)$
Convolution	$\sum_{m=0}^{N-1} s[m] \cdot u[n-m]$	$S(k) \cdot U(k)$
Time shifting	$s[n - m]$	$e^{-j \frac{2\pi k \cdot m}{T}} \cdot S(k)$
Frequency shifting	$e^{+j \frac{2\pi h n}{T}} \cdot s[n]$	$S(k - h)$

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References

1. Serge Lang, *Linear Algebra*, Springer Verlag New York Inc, 3rd Edition 1987.

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Schedule (tentative, visit regularly):

6-9	Organization and Introduction
13-9	Audio Production and Processing
20-9	ADC and an Algebraic Introduction to FT
27-9	FFT & FFT Workshop
4-10	Project Proposals (presentations by students)
11-10	Audio Features & workshop and data
18-10	Machine Learning
25-10	Student Paper Presentations I
1-11	Student Paper Presentations II
8-11	Student Paper Presentations III
15-11	Student Paper Presentations IV
22-11	No Class-Online Project Progress Meetings
29-11	Final Project Presentations Demo's
12-12	Project Deliverables: - Final Technical Project - Paper (4-8 pages), code, and - Web Site (or github)

Assignments (workshops):

1. [Vocal Tract Workshop](#). Due: September 20th 2022.
2. [FFT Workshop](#) and [audio_data](#). Due October 10th 2022.
3. Audio Features Workshop. Due 2022.
4. Machine Learning Workshop. Due 2022.

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API Project Proposals (October 4th 2022)

5 minute Presentations (4 slides) addressing:

- Title + group members (1 – 5 members)
- Problem description
- Challenges
- What will be the goal for the Final Project Presentation/Demo
- Note: If the group consists of more than 1 member, add a 5th slide with an initial global division of the work between project members. This slide does not have to be presented.

Each API Project member should submit a copy of the pdf with the slides of the API Project Proposal Presentation on Bright space before October 3rd 2022.

API Project Proposals (October 4th 2022)

For inspiration:

- See previous projects on <https://www.liacs.nl/~erwin/api>
- International Society for Music Information Retrieval (ISMIR)
<http://www.ismir.net/conferences/>
- INTERSPEECH
<https://www.isca-speech.org/iscaweb/index.php/online-archive>
- Online proceedings:
 - <https://dblp.org/db/conf/index.html>
 - <https://dblp.org/db/conf/interspeech/index.html>
 - <https://dblp.org/search?q=eurasip>
 - Etc.