

SHORT TIME FOURIER TRANSFORMS

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LML Audio Processing and Indexing

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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)

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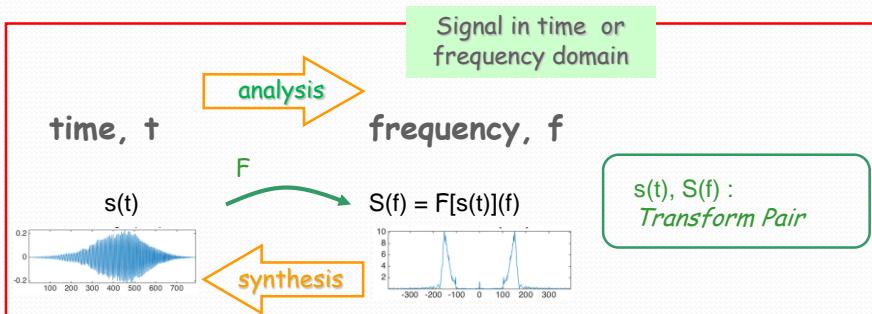
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Fourier Transforms

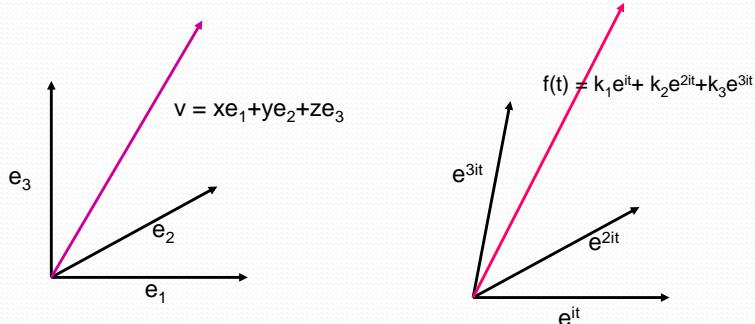
- Frequency analysis
- A tour of Fourier Transforms
- Continuous Fourier Series (FS)
- Discrete Fourier Series (DFS)

Frequency Analysis

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



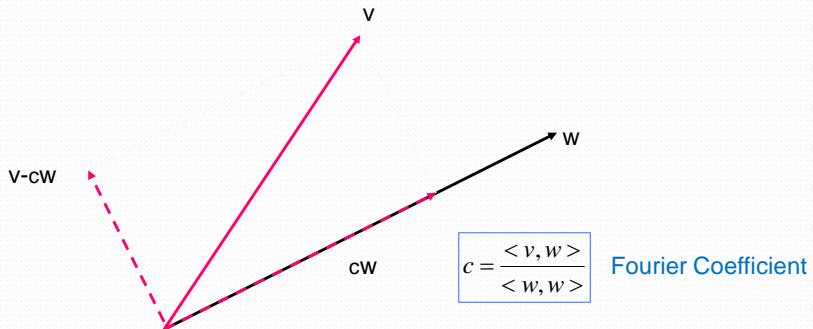
Bases of Vector Spaces



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:

$$c = \frac{< v, w >}{< w, w >}$$

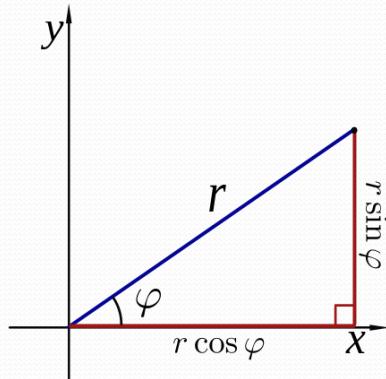
=> **cw** is the component of **v** along the direction of **w**.

Polar Coordinates in \mathbb{R}^2

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

$$x = r \cos \varphi$$

$$y = r \sin \varphi$$



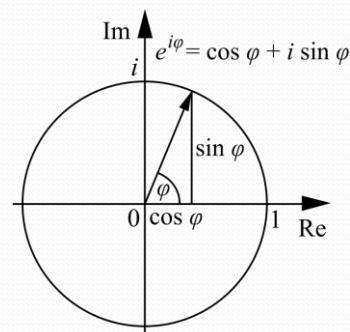
Complex Numbers

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

Note: you can write any complex number

$$z = a + bi$$
 as:

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



Complex Numbers and Functions

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

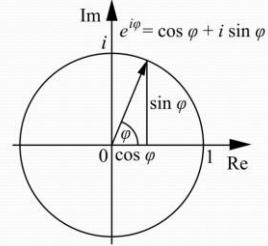
Let $z_1 = r_1 e^{-i\varphi_1}$, and $z_2 = r_2 e^{-i\varphi_2}$, then

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

Let f a given frequency.

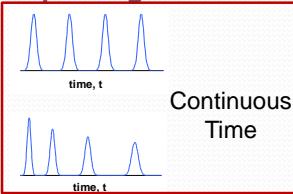
Let $h(t) = e^{i2\pi ft}$ then $h(t) = \cos 2\pi f t + i \sin 2\pi f t$, thus

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



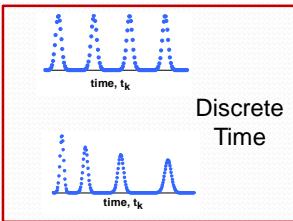
Fourier Analysis – Different 'Flavours'

Input Signal in Time Domain



Continuous Time

Periodic (period T) **FS**
 Aperiodic **FT**



Discrete Time

Periodic (period T) **DFS**
 Aperiodic **DTFT**
DFT ******

Frequency spectrum

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

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

$$\tilde{c}_k = \frac{1}{N} \sum_{n=0}^{N-1} s[n] \cdot e^{-\frac{i2\pi kn}{N}}$$

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

$$\tilde{c}_k = \frac{1}{N} \sum_{n=0}^{N-1} s[n] \cdot e^{-\frac{i2\pi kn}{N}}$$

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

** Calculated using FFT

History Fourier Transform

- **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

- **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).

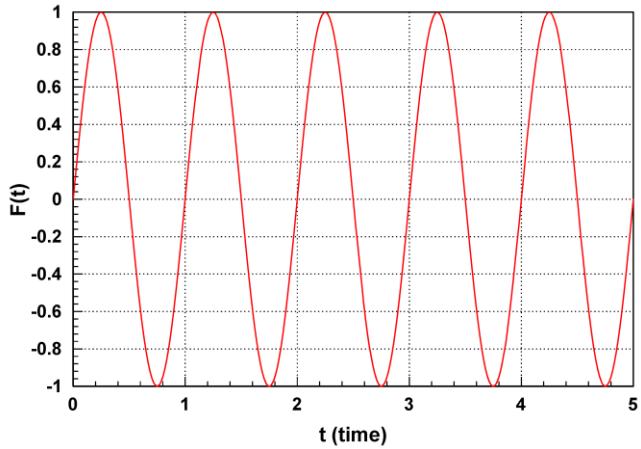
DISCRETE: Fast calculation methods (FFT)

- **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

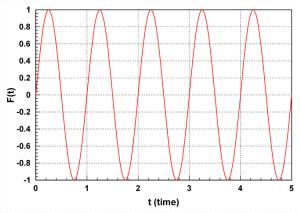


$$F(t) = \sin(2\pi t)$$

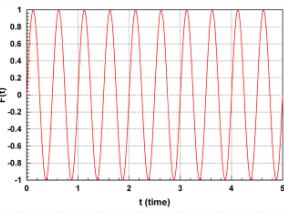
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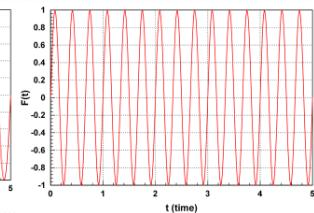
Another Space, Another Base



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



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



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

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

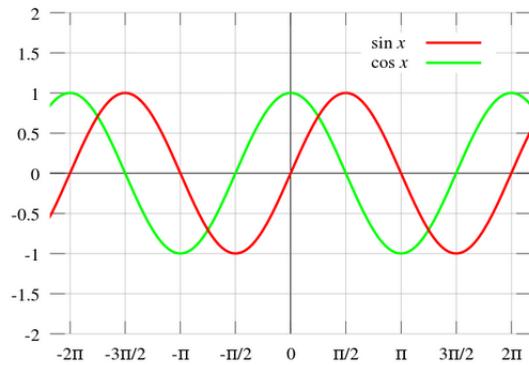
$$F(t) = \cos(2\pi \cdot 2t)$$

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

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

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Sine Cosine Graphs

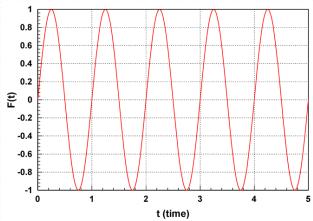


$$\sin(\varphi + \pi/2) = \cos(\varphi)$$

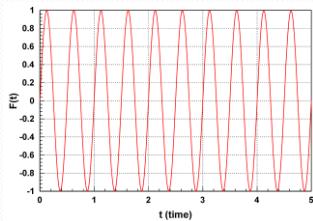
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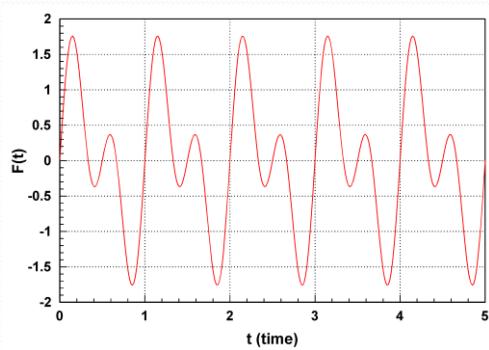
Linear Combination of Functions



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



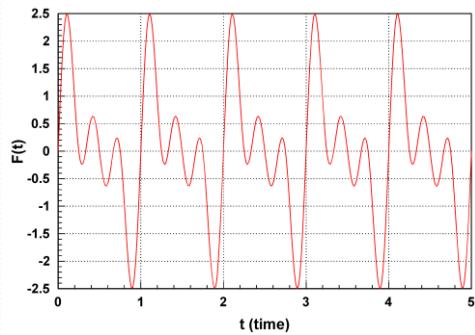
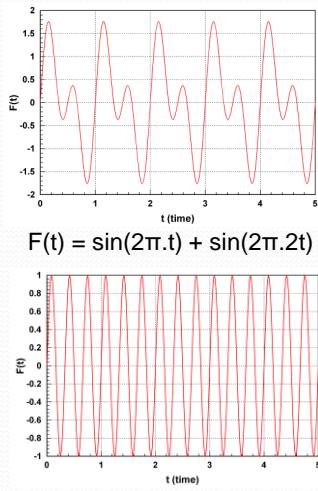
$$F(t) = \sin(2\pi.2t)$$



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

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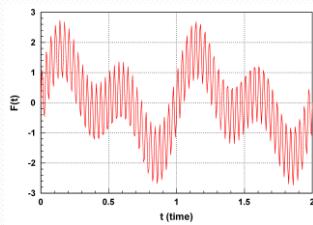
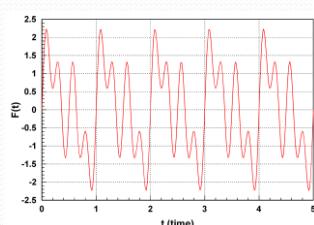
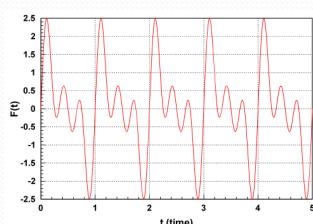
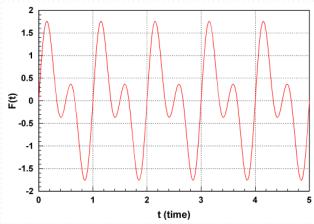
Linear Combination of Functions



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

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Linear Combination of Functions

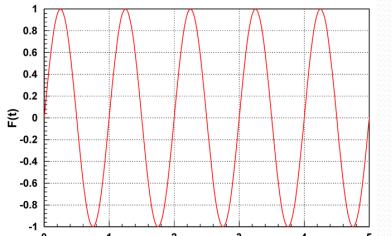


$F(t) = \sin(2\pi.t) + \sin(2\pi.2t) + \sin(2\pi.4t)$

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

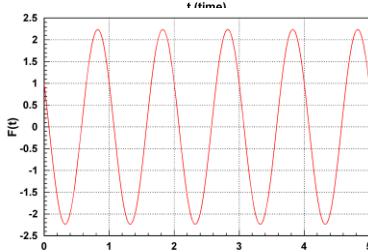
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Linear Combination of Functions



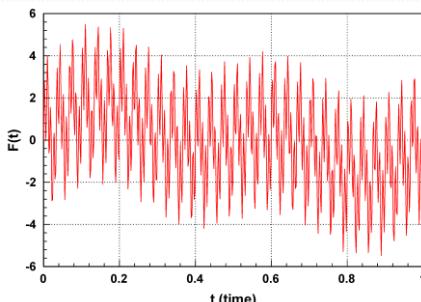
Phase Shift:

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



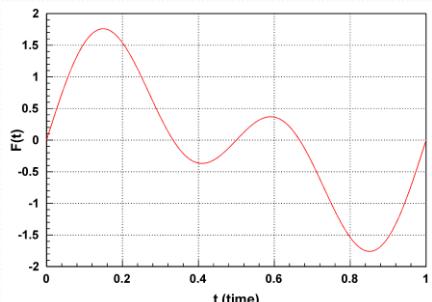
$$F(t) = \cos(2\pi \cdot t) - 2 \cdot \sin(2\pi \cdot t)$$

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.



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

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

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

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

Fourier Transform

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

$(a_0$ 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

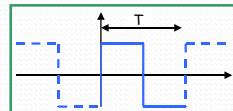
In any period:

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

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

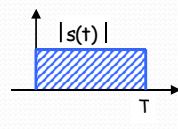
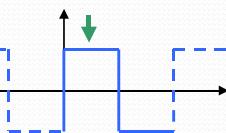
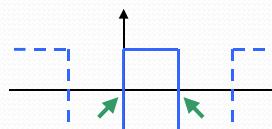
(c) $s(t)$ absolutely integrable, $\int_0^T |s(t)| dt < \infty$

Example:
square wave



Rate of convergence

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



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Fourier Series Analysis - 1

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

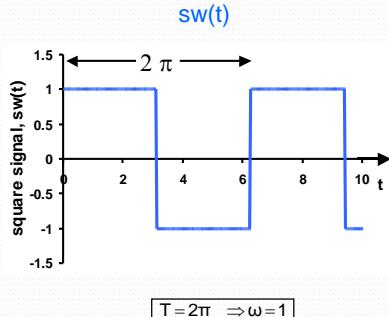
$$a_0 = \frac{1}{2\pi} \cdot \left\{ \int_0^{\pi} dt + \int_{\pi}^{2\pi} (-1)dt \right\} = 0 \quad (\text{zero average})$$

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

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

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

$$\begin{aligned} a_0 &= \frac{1}{T} \int_0^T s(t) dt \\ a_k &= \frac{2}{T} \int_0^T s(t) \cdot \cos(k\omega t) dt \\ -b_k &= \frac{2}{T} \int_0^T s(t) \cdot \sin(k\omega t) dt \end{aligned}$$



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

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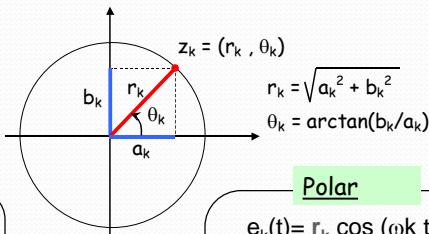
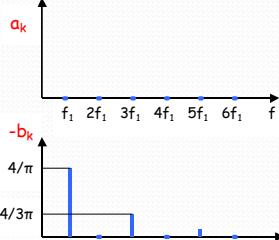
Fourier Series Analysis - 2

Fourier spectrum representations (in k)

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

Rectangular

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



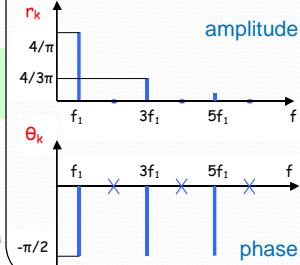
Polar

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

r_k = amplitude, θ_k = phase

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

Fourier spectrum of square-wave.



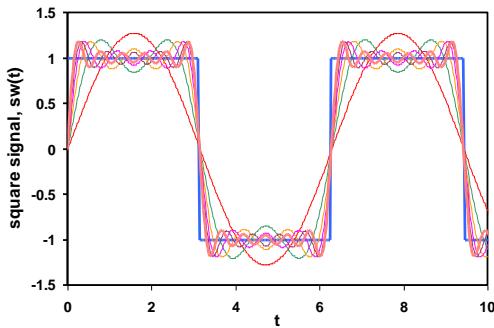
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Fourier Series Synthesis

Square wave reconstruction
from spectral terms

$$sw_7(t) = \sum_{k=1}^{79} [-b_k \cdot \sin(kt)]$$



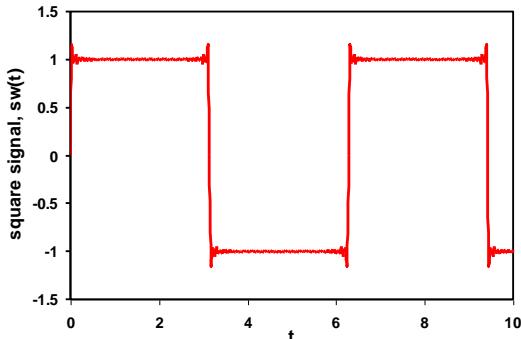
Convergence may be slow ($\sim 1/k$) - ideally need infinite terms.

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

Gibbs Phenomenon

Overshoot exist at each discontinuity

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



- 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*.

Fourier Series Time Shifting

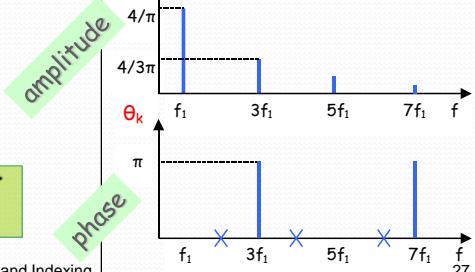
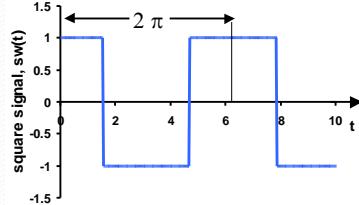
FS of even function:
 $\pi/2$ -advanced square-wave

$$a_0 = 0 \quad (\text{zero average})$$

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

$$-b_k = 0 \quad (\text{even function: } s(-x) = s(x))$$

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



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Complex Fourier Series

Euler's notation:

$$e^{-jt} = (e^{jt})^* = \cos(t) - i \cdot \sin(t) \quad \Rightarrow \text{"phasor"} \quad \cos(t) = \frac{e^{it} + e^{-it}}{2} \quad \sin(t) = \frac{e^{it} - e^{-it}}{2 \cdot i}$$

analysis

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

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

synthesis

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

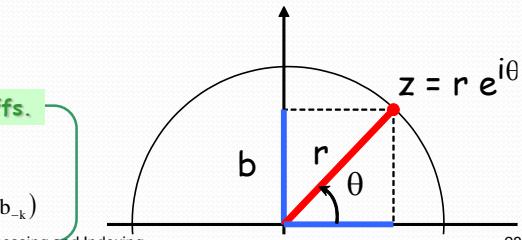
$$\text{Note: } c_{-k} = (c_k)^*$$

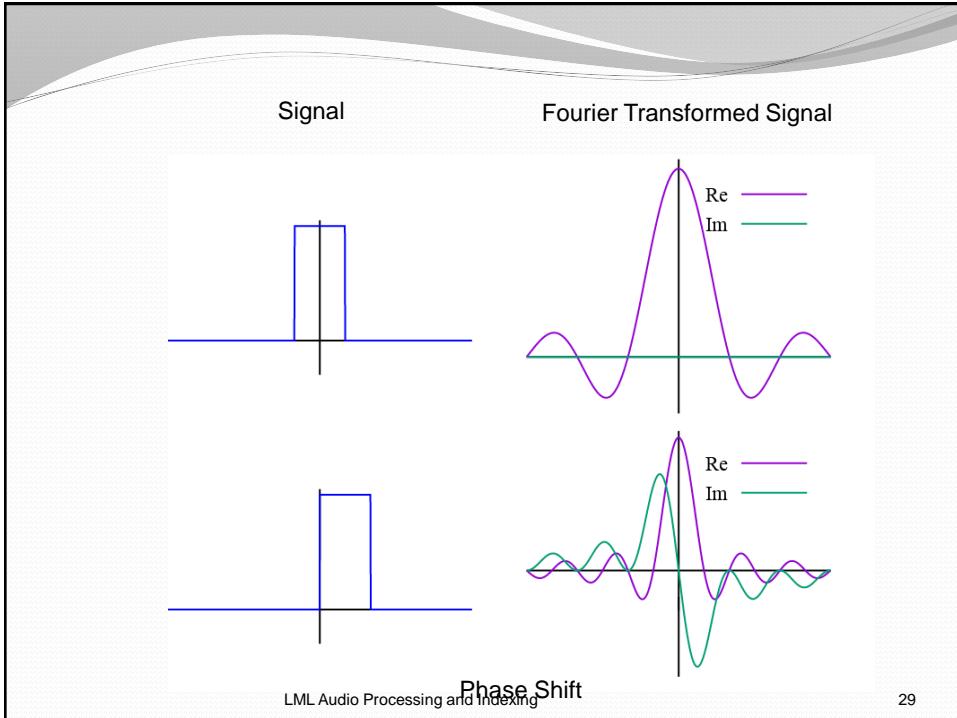
Link to FS real coeffs.

$$c_0 = a_0$$

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

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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	$s(t - \bar{t})$	$e^{-j \frac{2\pi f \cdot t}{T}} \cdot S(f)$
Frequency shifting	$e^{+j \frac{2\pi m t}{T}} \cdot s(t)$	$S(f - m)$

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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

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

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

IFT: synthesis

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

Orthogonality in DFS:

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

Kronecker's delta

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

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

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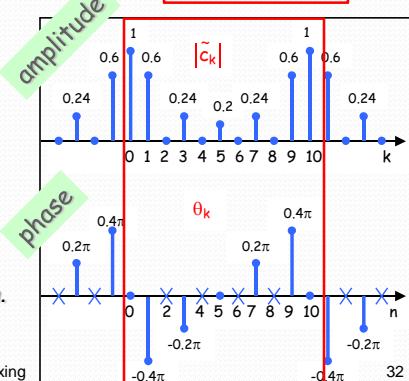
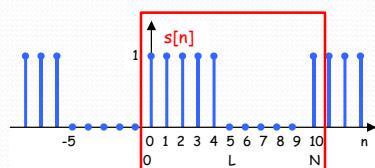
Discrete Fourier Series Analysis

DFS of periodic discrete 1-Volt square-wave

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

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

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	$s[n - m]$	$e^{-\frac{-i2\pi k \cdot m}{T}} \cdot S(k)$
Frequency shifting	$e^{\frac{+i2\pi h n}{T}} \cdot s[n]$	$S(k - h)$

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References

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

Schedule (tentative, visit regularly):

8-2	Organization and Introduction
15-2	Audio Production and Processing
22-2	ADC and an Algebraic Introduction to FT
1-3	No Class (See Brightspace)
4-3	Question hour: Project Proposals (See Brightspace 13.00 Kaltura)
8-3	Project Proposals (presentations by students)
15-3	FFT & FFI Workshop
22-3	Audio Features & workshop and data
29-3	Machine Learning + Workshop
5-4	Student Paper Presentations I.
12-4	Student Paper Presentations II.
19-4	Student Paper Presentations III.
26-4	Project Progress Reports
3-5	Team Meetings
10-5	Final Project Presentations Demo I
17-5	Final Project Presentations Demo II
23-5	Final Technical Project Paper (4-8 pages), code, and Web Site

Assignments (workshops):

1. [Vocal Tract Workshop](#). Due: 22-2 2022.
2. [FFT Workshop](#) and [audio_data](#). Due 22-3 2022 before 23.59.
3. Audio Features Workshop and data. Due 29-3 2022, 23.59.
4. Machine Learning Workshop. Due 8-4 2022, 23.59