

**6.003: Signals and Systems**

**Fourier Representations**

March 30, 2010

**Mid-term Examination #2**

Wednesday, April 7, 7:30-9:30pm, 34-101.

No recitations on the day of the exam.

Coverage: Lectures 1–15  
 Recitations 1–15  
 Homeworks 1–8

Homework 8 will not be collected or graded. Solutions will be posted.

Closed book: 2 pages of notes ( $8\frac{1}{2} \times 11$  inches; front and back).

Designed as 1-hour exam; two hours to complete.

Review sessions during open office hours.

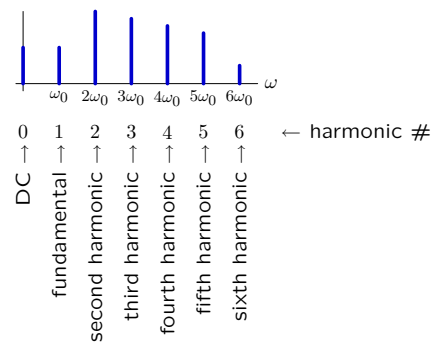
Conflict? Contact freeman@mit.edu before Friday, April 2, 5pm.

**Fourier Representations**

Fourier series represent **signals** in terms of **sinusoids**.  
 → leads to a new representation for **systems** as **filters**.

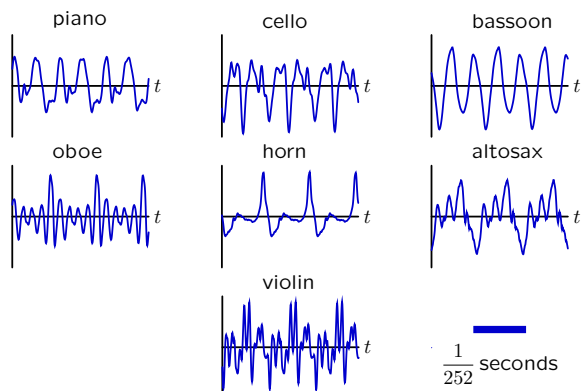
**Fourier Series**

Representing signals by their harmonic components.



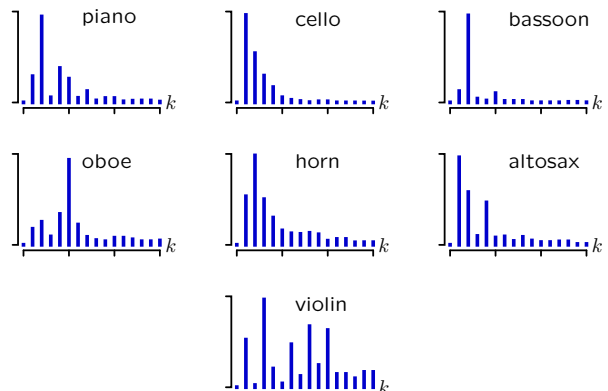
**Musical Instruments**

Harmonic content is natural way to describe some kinds of signals.  
 Ex: musical instruments (<http://theremin.music.uiowa.edu/MIS>)



**Musical Instruments**

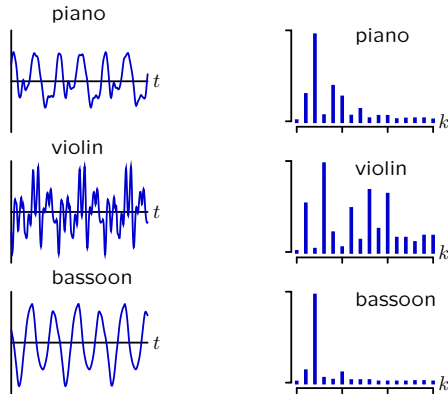
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**Musical Instruments**

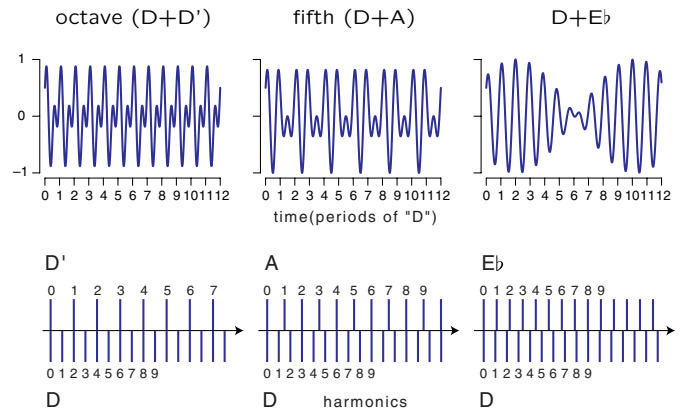
Harmonic content is natural way to describe some kinds of signals.

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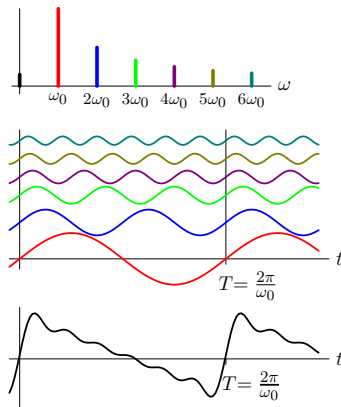
**Harmonics**

Harmonic structure determines consonance and dissonance.



**Harmonic Representations**

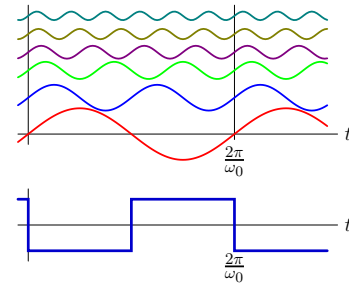
What signals can be represented by sums of harmonic components?



Only periodic signals: all harmonics of  $\omega_0$  are periodic in  $T = 2\pi/\omega_0$ .

**Harmonic Representations**

Is it possible to represent ALL periodic signals with harmonics? What about discontinuous signals?



Fourier claimed YES — even though all harmonics are continuous! Lagrange ridiculed the idea that a discontinuous signal could be written as a sum of continuous signals.

We will assume the answer is YES and see if the answer makes sense.

**Separating harmonic components**

Underlying properties.

1. Multiplying two harmonics produces a new harmonic with the same fundamental frequency:

$$e^{jk\omega_0 t} \times e^{jl\omega_0 t} = e^{j(k+l)\omega_0 t}$$

2. The integral of a harmonic over any time interval with length equal to a period  $T$  is zero unless the harmonic is at DC:

$$\int_{t_0}^{t_0+T} e^{jk\omega_0 t} dt \equiv \int_T e^{jk\omega_0 t} dt = \begin{cases} 0, & k \neq 0 \\ T, & k = 0 \end{cases} = T\delta[k]$$

**Separating harmonic components**

Assume that  $x(t)$  is periodic in  $T$  and is composed of a weighted sum of harmonics of  $\omega_0 = 2\pi/T$ .

$$x(t) = x(t + T) = \sum_{k=-\infty}^{\infty} a_k e^{j\omega_0 k t}$$

Then

$$\begin{aligned} \int_T x(t) e^{-jl\omega_0 t} dt &= \int_T \sum_{k=-\infty}^{\infty} a_k e^{j\omega_0 k t} e^{-j\omega_0 l t} dt \\ &= \sum_{k=-\infty}^{\infty} a_k \int_T e^{j\omega_0 (k-l)t} dt \\ &= \sum_{k=-\infty}^{\infty} a_k T \delta[k-l] = T a_l \end{aligned}$$

Therefore

$$a_k = \frac{1}{T} \int_T x(t) e^{-j\omega_0 k t} dt = \frac{1}{T} \int_T x(t) e^{-j\frac{2\pi}{T} k t} dt$$

**Fourier Series**

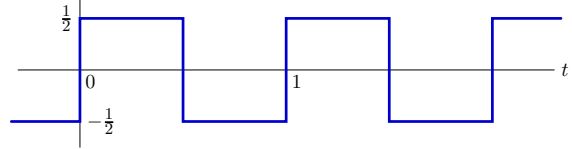
Determining harmonic components of a periodic signal.

$$a_k = \frac{1}{T} \int_T x(t) e^{-j\frac{2\pi}{T}kt} dt \quad (\text{"analysis" equation})$$

$$x(t) = x(t + T) = \sum_{k=-\infty}^{\infty} a_k e^{j\frac{2\pi}{T}kt} \quad (\text{"synthesis" equation})$$

**Check Yourself**

Let  $a_k$  represent the Fourier series coefficients of the following square wave.



How many of the following statements are true?

1.  $a_k = 0$  if  $k$  is even
2.  $a_k$  is real-valued
3.  $|a_k|$  decreases with  $k^2$
4. there are an infinite number of non-zero  $a_k$
5. all of the above

**Fourier Series Properties**

If a signal is differentiated in time, its Fourier coefficients are multiplied by  $j\frac{2\pi}{T}k$ .

Proof: Let

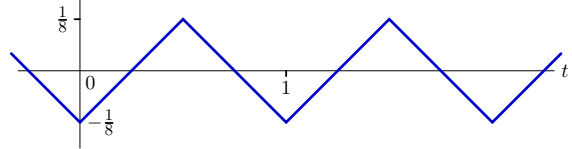
$$x(t) = x(t + T) = \sum_{k=-\infty}^{\infty} a_k e^{j\frac{2\pi}{T}kt}$$

then

$$\dot{x}(t) = \dot{x}(t + T) = \sum_{k=-\infty}^{\infty} \left( j\frac{2\pi}{T}k a_k \right) e^{j\frac{2\pi}{T}kt}$$

**Check Yourself**

Let  $b_k$  represent the Fourier series coefficients of the following triangle wave.



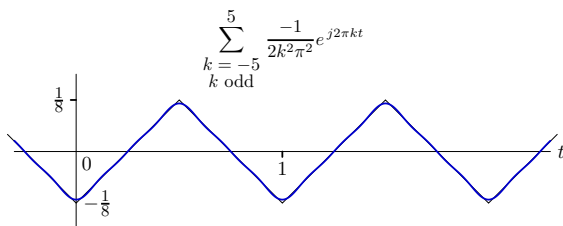
How many of the following statements are true?

1.  $b_k = 0$  if  $k$  is even
2.  $b_k$  is real-valued
3.  $|b_k|$  decreases with  $k^2$
4. there are an infinite number of non-zero  $b_k$
5. all of the above

**Fourier Series**

One can visualize convergence of the Fourier Series by incrementally adding terms.

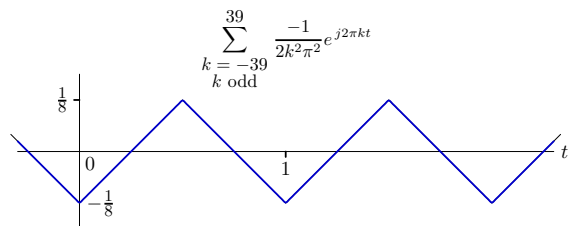
Example: triangle waveform



**Fourier Series**

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Example: triangle waveform

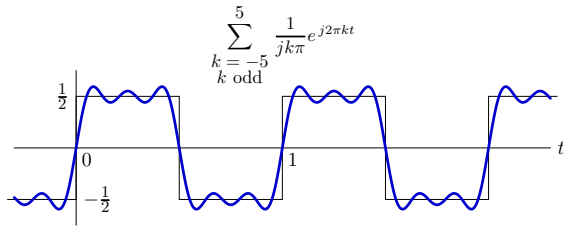


Fourier series representations of functions with discontinuous slopes converge toward functions with discontinuous slopes.

**Fourier Series**

One can visualize convergence of the Fourier Series by incrementally adding terms.

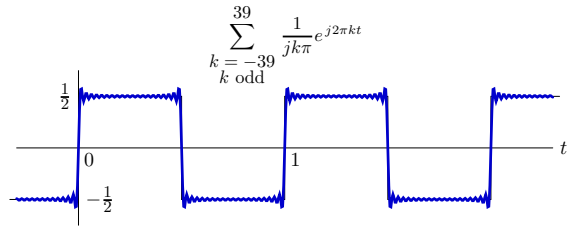
Example: square wave



**Fourier Series**

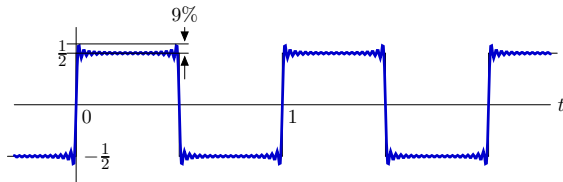
One can visualize convergence of the Fourier Series by incrementally adding terms.

Example: square wave



**Fourier Series**

Partial sums of Fourier series of discontinuous functions “ring” near discontinuities: Gibb’s phenomenon.



This ringing results because the magnitude of the Fourier coefficients is only decreasing as  $\frac{1}{k}$  (while they decreased as  $\frac{1}{k^2}$  for the triangle).

You can decrease (and even eliminate the ringing) by decreasing the magnitudes of the Fourier coefficients at higher frequencies.

**Fourier Series: Summary**

Fourier series represent periodic signals as sums of sinusoids.

- valid for an extremely large class of periodic signals
- valid even for discontinuous signals such as square wave

However, convergence as # harmonics increases can be complicated.

**Filtering**

The output of an LTI system is a “filtered” version of the input.

Input: Fourier series → sum of complex exponentials.

$$x(t) = x(t + T) = \sum_{k=-\infty}^{\infty} a_k e^{j\frac{2\pi}{T}kt}$$

Complex exponentials: eigenfunctions of LTI systems.

$$e^{j\frac{2\pi}{T}kt} \rightarrow H(j\frac{2\pi}{T}k)e^{j\frac{2\pi}{T}kt}$$

Output: same eigenfunctions, amplitudes/phases set by system.

$$x(t) = \sum_{k=-\infty}^{\infty} a_k e^{j\frac{2\pi}{T}kt} \rightarrow y(t) = \sum_{k=-\infty}^{\infty} a_k H(j\frac{2\pi}{T}k)e^{j\frac{2\pi}{T}kt}$$

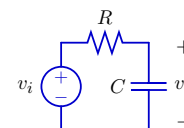
**Filtering**

Notion of a filter.

LTI systems

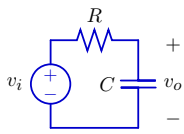
- cannot create new frequencies.
- can scale magnitudes and shift phases of existing components.

Example: Low-Pass Filtering with an RC circuit

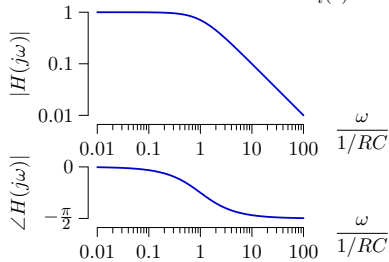


**Lowpass Filter**

Calculate the frequency response of an RC circuit.

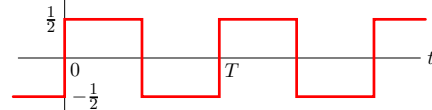


KVL:  $v_i(t) = Ri(t) + v_o(t)$   
 C:  $i(t) = C\dot{v}_o(t)$   
 Solving:  $v_i(t) = RC\dot{v}_o(t) + v_o(t)$   
 $V_i(s) = (1 + sRC)V_o(s)$   
 $H(s) = \frac{V_o(s)}{V_i(s)} = \frac{1}{1 + sRC}$

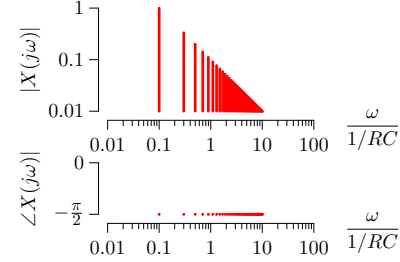


**Lowpass Filtering**

Let the input be a square wave.

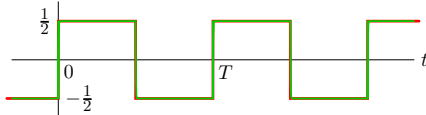


$$x(t) = \sum_{k \text{ odd}} \frac{1}{j\pi k} e^{j\omega_0 kt}; \quad \omega_0 = \frac{2\pi}{T}$$

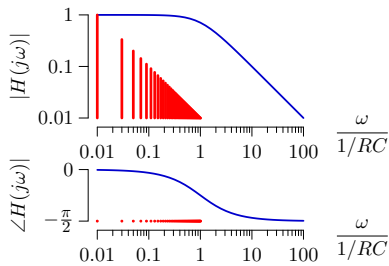


**Lowpass Filtering**

Low frequency square wave:  $\omega_0 \ll 1/RC$ .

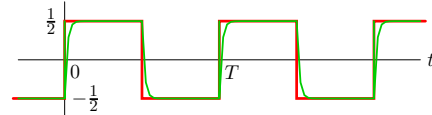


$$x(t) = \sum_{k \text{ odd}} \frac{1}{j\pi k} e^{j\omega_0 kt}; \quad \omega_0 = \frac{2\pi}{T}$$

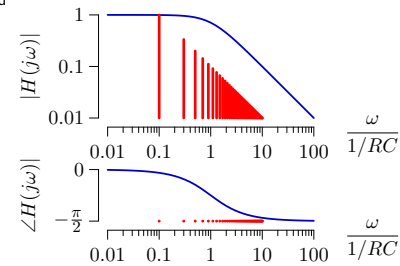


**Lowpass Filtering**

Higher frequency square wave:  $\omega_0 < 1/RC$ .

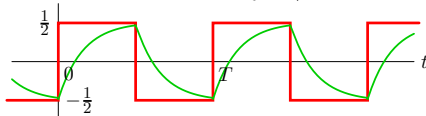


$$x(t) = \sum_{k \text{ odd}} \frac{1}{j\pi k} e^{j\omega_0 kt}; \quad \omega_0 = \frac{2\pi}{T}$$

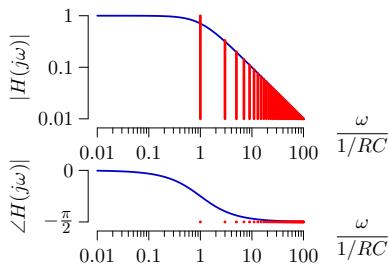


**Lowpass Filtering**

Still higher frequency square wave:  $\omega_0 = 1/RC$ .

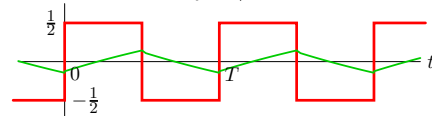


$$x(t) = \sum_{k \text{ odd}} \frac{1}{j\pi k} e^{j\omega_0 kt}; \quad \omega_0 = \frac{2\pi}{T}$$

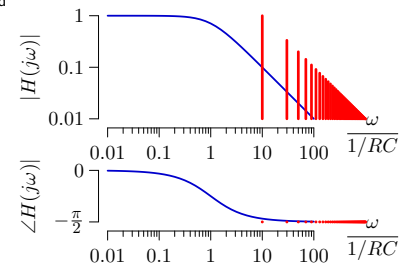


**Lowpass Filtering**

High frequency square wave:  $\omega_0 > 1/RC$ .



$$x(t) = \sum_{k \text{ odd}} \frac{1}{j\pi k} e^{j\omega_0 kt}; \quad \omega_0 = \frac{2\pi}{T}$$



**Fourier Series: Summary**

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Fourier series represent signals by their frequency content.

Representing a signal by its frequency content is useful for many signals, e.g., music.

Fourier series motivate a new representation of a system as a filter.