

Note: by the time-shift result

of assignment 2, and superposition,

the coefficients of the Fourier

series of the output of the full-wave

rectifier should be

$$c_n = \left[1 + e^{j \frac{2\pi}{T} n \Delta} \right] \times \text{coefft of } \frac{1}{2}\text{-wave series.}$$

where $\Delta = \frac{T}{2}$.

This gives,

$$\text{for } n=0 \quad - \quad c_n = 2$$

$$\text{for } n=\pm 1 \quad - \quad c_n = \left(1 + e^{\pm j\pi} \right) \cdot \frac{A}{4j} = 0$$

$$\text{for } |n| \geq 2, \quad \begin{aligned} c_n &= \left(1 + e^{j\pi n} \right) \frac{-A}{\pi(n^2-1)} \\ &= \frac{-2A}{\pi(n^2-1)} \end{aligned}$$

$$\text{for } |n| \geq 2, \quad \begin{aligned} |n| \text{ odd} \quad \rightarrow c_n &= 0 \end{aligned}$$

- This agrees with our direct calculation.

We motivated the use
of Fourier series by pointing
out that it would let us
extend our frequency-response
analysis to periodic signals
other than pure sinusoids.

To illustrate, suppose
that the Fourier series

$$\sum_{n=-\infty}^{\infty} c_n e^{j \frac{2\pi}{T} n t}$$

represents the output of a
rectifier, and that we now
wish to filter that signal
— perhaps to produce a
dc power supply.

Suppose that the filter has a transfer function $H(s)$. Then its response to each of the sinusoids in the Fourier series is given by its frequency response; and by superposition, its response to the entire Fourier series is

$$\sum_{n=-\infty}^{\infty} c_n H(j \frac{2\pi}{T} n) e^{j \frac{2\pi}{T} n t}$$

$$= \sum_{n=-\infty}^{\infty} c_n |H(j \frac{2\pi}{T} n)| e^{j(\frac{2\pi}{T} n t + \angle H(j \frac{2\pi}{T} n))}$$

— so we can easily find the output of the filter, in the form of a Fourier series.

We can then quantify
the effect of the filtering in
terms of amplitudes, or,
using Parseval's Theorem, in
terms of average power.

Example: Let's try filtering the output of the full-wave rectifier.

Call the input frequency ω_0 .

We wish to pass the dc component, but block the first harmonic of the input frequency at $2\omega_0$.

(Recall that the full-wave rectifier output has no frequency component at ω_0 itself.)

The dc component is $\frac{2}{\pi} A$,

so its power is $\frac{4}{\pi^2} A^2$.

The component at the frequency $2\omega_0$ has an amplitude of

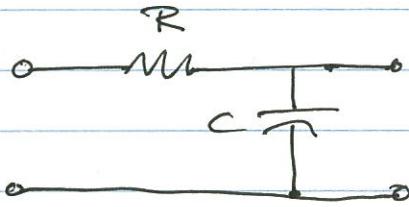
$$\frac{4}{\pi} A \cdot \frac{1}{3},$$

so its power is $\frac{8}{9 \pi^2} A^2$.

We need to reduce the power of the signal at $2\omega_0$ by a factor of about 25, or $10 \log 25 \text{ dB} = 14 \text{ dB}$.

Let's be conservative and go for 20 dB.

If we use the following simple first-order filter:



Then the frequency $2\omega_0$ will have to be one decade above the "corner frequency" of the filter.

→ Place corner frequency at

$$\frac{2\omega_0}{10} \Leftrightarrow \gamma = RC = \frac{5}{\omega_0}$$

So the dc component of
the filter output is

$$\frac{2A}{\pi} H(0) = \frac{2A}{\pi}$$

... and the amplitude of
the component at the frequency
 $2\omega_0$ is

$$\frac{4A}{\pi} \cdot \frac{1}{3} \left| \frac{1}{1+j5.2} \right|$$

$$\sim \frac{4}{3\pi} \cdot \frac{1}{10.5} A$$

$$\sim 0.04 A$$

The power in the dc component is therefore

$$\frac{4 A^2}{\pi^2}$$

while the power in the 2 w_o component is

$$\frac{1}{2} \cdot 16 \times 10^{-4} \text{ A}^2$$

$$\sim 8 \times 10^{-4} \text{ A}^2$$

Real sinusoidal form of Fourier series:

Suppose $f(t)$ is real-valued.

Then, for any n ,

$$c_{-n} = c_n^*$$

It follows that the Fourier series

is

$$\sum_{n=-\infty}^{\infty} c_n e^{j \frac{2\pi}{T} nt}$$

$$= c_0 + \sum_{n=1}^{\infty} (c_n e^{j \frac{2\pi}{T} nt} + c_n^* e^{-j \frac{2\pi}{T} nt})$$

$$= c_0 + \sum_{n=1}^{\infty} 2 \operatorname{Re} (c_n e^{j \frac{2\pi}{T} nt})$$

$$= c_0 + \sum_{n=1}^{\infty} 2 \operatorname{Re} (|c_n| e^{j(\frac{2\pi}{T} nt + \angle c_n)})$$

$$= c_0 + \sum_{n=1}^{\infty} 2 |c_n| \cos \left(\frac{2\pi}{T} nt + \angle c_n \right)$$

But this is

$$c_0 + \sum_{n=1}^{\infty} 2|c_n| \left(\cos \angle c_n \cos \frac{2\pi}{T} nt - \sin \angle c_n \sin \frac{2\pi}{T} nt \right)$$

$$= a_0 + \sum_{n=1}^{\infty} \left(a_n \cos \frac{2\pi}{T} nt + b_n \sin \frac{2\pi}{T} nt \right),$$

where

$$a_0 = c_0,$$

and for $n \geq 1$,

$$a_n = 2 \operatorname{Re} c_n$$

$$b_n = -2 \operatorname{Im} c_n$$

It follows that

$$a_n = 2 \operatorname{Re} \left[\frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} f(t) e^{-j \frac{2\pi}{T} nt} dt \right]$$

$$= \frac{2}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} f(t) \cos \frac{2\pi}{T} nt dt$$

$$\& b_n = -2 \operatorname{Im} \left[\frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} f(t) e^{-j \frac{2\pi}{T} nt} dt \right]$$

$$= \frac{2}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} f(t) \sin \frac{2\pi n}{T} t dt$$

$$(\& a_0 = \frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} f(t) dt).$$

Gibbs' phenomenon

The partial sums of the square-wave Fourier series "overshoot" the square wave itself in the region of the discontinuities.

This overshoot doesn't tend to zero as the number of terms goes to infinity - it tends to something like 9% of the "jump."

This may seem surprising in light of the Dirichlet convergence theorem, but it is, of course, consistent.

In fact, the same thing happens at discontinuities of any $f(t)$.

Though the Fourier series does not converge pointwise to $f(t)$ at every point

(namely, not at discontinuities),

the average power of the approximation error of the partial sums tends to zero

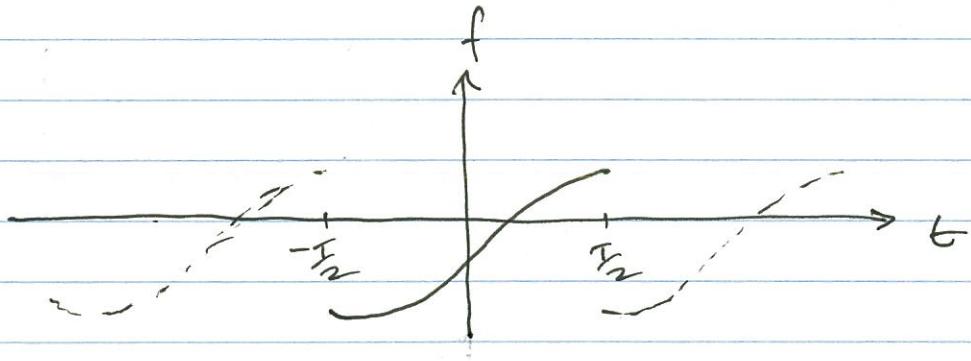
— we say that the

Fourier series "converges in mean square" to $f(t)$.

The Fourier transform

Suppose we wished to extend the Fourier series to functions that aren't periodic.

We can always approximate $f(t)$ over an interval $[-\frac{T}{2}, \frac{T}{2}]$ by pretending that it's periodic with period T .



This gives a Fourier series with frequency components at integer multiples of $\frac{2\pi}{T}$.

Call the frequency increment

$$\Delta \omega = \frac{2\pi}{T}$$

Then

$$f(t) \sim \sum_{n=-\infty}^{\infty} c_n e^{j \Delta \omega n t}$$
$$= \frac{1}{2\pi} \sum_{n=-\infty}^{\infty} (T c_n) e^{j \Delta \omega n t}$$

Now, as $T \rightarrow \infty$ this sum

should approach an integral

$$\frac{1}{2\pi} \int_{-\infty}^{\infty} \hat{f}(\omega) e^{j \omega t} dt,$$

where $\hat{f}(\omega)$ is what $T c_n$ approaches as $T \rightarrow \infty$; namely,

$$\hat{f}(\omega) = \frac{1}{T} \int_{-\frac{T}{2}}^{\frac{T}{2}} f(t) e^{-j\omega t} dt$$

We call

$$\mathcal{F}\{f\} = \hat{f}(\omega)$$

the Fourier transform of $f(t)$,

and we call the integral

$$f(t) = \frac{1}{2\pi} \int_{-\infty}^{\infty} \hat{f}(\omega) e^{j\omega t} d\omega$$

the Fourier transform inversion

integral.

Note that the Fourier transform
resembles the Laplace transform
with $s = j\omega$.

Indeed, it has similar
properties with respect to time-
shifts, sinusoidal modulation,
convolution and the like,
as long as one restricts attention
to functions satisfying
 $\int_{-\infty}^{\infty} |f(t)| dt < \infty$.

It also satisfies a version
of Parseval's Theorem:

$$\int_{-\infty}^{\infty} |f(t)|^2 dt = \frac{1}{2\pi} \int_{-\infty}^{\infty} |\hat{f}(\omega)|^2 d\omega$$