Fourier Transforms

ECE 320 Electronics I

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Please visit Bison Academy for correspondinglecture notes, homework sets, and solutions

Problem

- Find $V2(t)$ given $V1(t)$
- Previous lectures simplified thisproblem
	- Assume V1(t) is a sine wave
	- Results are close but slightly off

You *can* solve for V2(t) exactly

- Much harder solution
- For slightly more accurate answers

Solution uses Fourier Transforms

• Once you have phasors and superposition, you have Fourier Transforms

-0.7V

1ms

Phasors

- Real numbers are all that's needed for DC analysis
- Complex numbers help with AC analysis

Voltages

 $v(t) = a \cos(\omega t) + b \sin(\omega t)$ *time domain* $V = a - ib$ ⁼ *^a* [−] *jb ^phasor (frequency) domain*

Impedances

 $R \rightarrow R$
r $L \rightarrow j\omega L$
 ≈ 1 $C \rightarrow \frac{1}{j\omega C}$

Voltage nodes, current loops, voltage division etc. work for both DC and AC

• You get complex numbers with AC however

Superposition

Linear Systems

f(*a* + *b*) = *f*(*a*) + *f*(*b*)

If there are several inputs

- Analyze separately for each input .
- The total input is found by summing up each of the inputs.
- The total output is found by summing up each of the outputs.

Fourier Transforms convert a signal that is *not* a bunch of sine waves into a signalwhich *is* a bunch of sine waves

Allows you to solve using phasor analysis and superposition

Fourier Transform

Assume a signal is periodic in time T:

$$
x(t) = x(t+T)
$$

then

$$
x(t) = \sum_{n=0}^{\infty} a_n \cos(n\omega_0 t) + b_n \sin(n\omega_0 t) \qquad \omega_0 = \frac{2\pi}{T}
$$

Translation:

If you add up a bunch of signals which are periodic in time T, the result is periodic in time TIf you have a periodic signal which is not a pure sine wave, it is made up of harmonics.

Computing Fourier Coefficients

All sine waves are orthogonal $mean(\sin(at) \cdot \cos(bt)) = 0$ *mean* $n(\sin$ $(at) \cdot \sin(bt)) =$ \int $\bigg)$ $\bigg\{$ 0*a*≠*b* $\frac{1}{2}$ $a =$ *a*=*bmean* $n(\cos(at) \cdot \cos(bt)) =$ \int $\bigg)$ $\bigg\{$ 0*a*≠*b* $\frac{1}{2}$ $a=b$

meaning:

 $a_0 = mean(x(t))$ the DC value of $x(t)$

 $a_n = 2 \cdot mean(x(t) \cdot \cos(n\omega_0 t))$ cosine() terms

$$
b_n = 2 \cdot mean(x(t) \cdot \sin(n\omega_0 t))
$$

since() terms

Example 1: Known answer

 $x(t) = 1 + 3\cos(t) + 4\sin(2t)$

In Matlab

```
t = [1:10000]' / 10000 * 2 * pi;
x = 1 + 3 \times \cos(t) + 4 \times \sin(2 \times t);
a0 = \text{mean}(x)
a0 = 1.0000a1 = 2*mean(x : x cos(t))a1 = 3.0000b1 = 2*mean(x : * sin(t))b1 = 2.9165e-015a2 = 2*mean(x \cdot * cos(2*t))a2 = -3.0340e - 015b2 = 2*mean(x : * sin(2*t))b2 = 4.0000
```


Complex Fourier Transform

Easier if you don't mind complex numbers

Xn=*an*− *jb* $n = 2 \cdot mean(x(t) \cdot e^{-jn})$ ω0*t*)

```
X1 = 2*mean(x \cdot * exp(-i*t))X1 = 3.0000 - 0.0000iX2 = 2*mean(x, * exp(-i * 2 * t))X2 = -0.0000 - 4.0000iX3 = 2*mean(x \cdot * exp(-i*3*t))
X3 = 5.4526e-015 +5.7784e-016i
```
Example: Buck Converter

Previous Solution

Approximate V1(t) as

- A DC term (7.58V), and
- An AC term $(20.7Vpp \ @$ 1kHz)

The answers we got for the voltage at V2 were close to what CircutLab computed,but a little off.

Using Fourier Transforms, you can getmore accurate results.

1) Find the Fourier Series expansion for V1(t).

- Change in variable so the period is 2π
- Has no affect on the Fourier coefficients

```
t = [1:10000]' / 10000;
x = 20*( t < 0.4) - 0.7*( t >= 0.4);t = t * 2 * pi;X0 = \text{mean}(x)
X0 = 7.5779X1 = 2*mean(x, * exp(-i*t))X1 = 3.8725 -11.9184iX2 = 2*mean(x \cdot * exp(-i * 2 * t))X2 = -3.1360 - 2.2784iX3 = 2*mean(x : x \text{ exp}(-j*3*t))X3 = 2.0861 - 1.5157iX4 = 2*mean(x \cdot * exp(-i * 4 * t))X4 = -0.9686 - 2.9811i
```


What this means is that

```
V1 = 7.5779
 + 2.8725 cos(t) + 11.9184 sin(t)
 - 3.1360 cos(2*t) + 2.2784 sin(2*t)+ 2.0861 \cos(3*t) + 1.515 \sin(3*t)
 - 0.9686 cos(4*t) + 2.9811 sin(4*t)
```
• Note: Adding more terms improves the approximation.

Now use superposition to solve for $V2(t)$

- Treat this as 5 separate problems: each at a different frequency
- Solve for $V2(t)$ at each frequency
- Add up the answers to get the total answer.

DC Analysis: $V_1(t) = 7.5779$

Redraw the circuit at $w = 0$ and solve for V2

$$
V_1 = 7.5779 + j0
$$

$$
V_2 = \left(\frac{100}{100 + 15}\right) 7.5779
$$

 V_2 = 6.5895

1st Harmonic: 1000Hz

$$
V_1 = 2.8725 \cos(\omega_0 t) + 11.9184 \sin(\omega_0 t) \rightarrow 2.8725 - j11.9184
$$

\n
$$
V_2 = \left(\frac{(2.4705 - j15.5223)}{(2.4705 - j15.5223) + (15 + j628)}\right) (2.8725 - j11.9184)
$$

\n
$$
V_2 = -0.1290 + j0.2866
$$

2nd Harmonic (2000Hz)

*V*2 ⁼ 0.0185 ⁺ *^j*0.0162

 $V_1 = -3.1360 \cos(2\omega_0 t) + 2.2784 \sin(2\omega_0 t) \rightarrow -3.1360 - j2.2784$ $V_2 = \left(\frac{(0.6293 - j7.9077)}{(0.6293 - j7.9077) + (15 + j1256.6)}\right) (-3.1360 - j2.2784)$

+
\
- $\begin{array}{c|c|c} \begin{matrix} \n\text{V1} & \text{V1} & \text{V2} & \text{V2} \ \n\end{matrix} & & \begin{matrix} \text{V2} & \text{V1} & \text{V2} & \text{V3} \ \text{V4} & \text{V5} & \text{V6} & \text{V7} \ \end{matrix} & & \begin{matrix} \text{V1} & \text{V2} & \text{V4} & \text{V5} \ \text{V1} & \text{V2} & \text{V6} & \text{V7} \ \end{matrix} & & \begin{matrix} \text{V1} & \text{V2} & \text{V6} & \text{$ 15 **j1256.6 -j7.9577-3.1360 - j2.2784**0.6293 - j7.90772000Hz

3rd Harmonic: 3000Hz

$$
V_1 = 2.0861 \cos(3\omega_0 t) + 1.5150 \sin(3\omega_0 t) \rightarrow 2.0861 - j1.5150
$$

$$
V_2 = \left(\frac{(0.2807 - j5.2903)}{(0.2807 - j5.2903) + (15 + j1884.9)}\right) (2.0861 - j1.5150)
$$

$$
V_2 = -0.00061 + j0.00039
$$

4th Harmonics: 4000 Hz

$$
V_1 = -0.9684 \cos (4\omega_0 t) + 2.9811 \sin (4\omega_0 t) \rightarrow -0.9684 - j2.9811
$$

$$
V_2 = \left(\frac{(0.1582 - j3.9726)}{(0.1582 - j3.9726) + (15 + j2513.27)}\right) (-0.9684 - j2.9811)
$$

$$
V_2 = 0.00132 + j0.00479
$$

Net Result:

 $V_2 = 6.5895 +$ $-0.1290\cos{(\omega_0 t)} - 0.2866\sin{(\omega_0 t)}$ $+0.0185\cos(2\omega_0 t) - 0.0162\sin(2\omega_0 t)$ $-0.00061 \cos(3\omega_0 t) - 0.00039 \sin(3\omega_0 t)$ $+0.00132\cos(4\omega_0t) - 0.00479\sin(4\omega_0t)$ $+\cdots$

Note

- In theory, you have to go out to infinity
- In practice, the harmonics go to zero very quickly
- Truncating the series after the 4th harmonic isvery close

In Matlab:

- Output $=$ Gain $*$ Input
- note: slightly different answers from before due to rounding
- $>> \text{Y0} = 100/(100+15)$ *X0 6.5895
- >> Y1 = $(2.4705 j*15.52) / ((2.4705 j*15.52) + (15 + j*628.2)) * X1$ $-0.1542 + 0.2819i$
- \geq Y2 = (0.6293-j*7.9077)/((0.6293-j*7.9077)+(15+j*1256.6))*X2 0.0185 + 0.0162i
- >> Y3 = $(0.2807 j*5.2903) / ((0.2807 j*5.2903) + (15 + j*1884.9)) * X3$ $-0.0061 + 0.0039i$
- >> Y4 = $(0.1582 j*3.9726) / ((0.1582 j*3.9726) + (15 + j*2513.27)) * X4$ 0.0013 + 0.0048i

Plotting V2(t):

```
V2 = Y0 + real(Y1) * cos(t) - imag(Y1) * sin(t);
V2 = V2 + real(Y2)*cos(2*t) - imag(Y2)*sin(2*t);
V2 = V2 + real(Y3)*cos(3*t) - imag(Y3)*sin(3*t);
V2 = V2 + real(Y4)*cos(4*t) - imag(Y4)*sin(4*t);plot(t,V2)\gt
```
This matches CircuitLab results

- In theory, you need to include an infinite number of terms
- In practice, the terms quickly go to zero
- Fourier Transforms allow you to compute theexplicit form for $V2(t)$,
- Fourier Transforms are more accurate than whatwe did last lecture, and
- They are a *lot* more work.

Handout

Assume the Fourier transform for $V1(t)$ is

 $V1(t) = 113.78 - 6.26 \cos(6280t) + 8.62 \sin(6280t)$ Find V2(t)

Sidelight:

Previous lectures approxmated V1(t) as

- DC term (mean of $V1$)
- AC term $(20.7Vpp \ @$ 1kHz)

A more accurate approximation is

- DC Term (mean of V1)
- AC Term (25.06Vpp @ 1kHz) 1st term of Fourier series)

DC Analysis in Matlab:

 $V1dc = mean(x)$ 7.5779

```
V2dc = 100/(100+15) * V1dc 6.5895
```
This matches

- CircuitLab
- The result using Fourier transforms out to the4th harmonic

AC Analysis in Matlab

- $X1 = 2*mean(x, * exp(-j*t))$ 3.8725 -11.9184i
- $V1pp = 2*abs(X1)$ 25.0635 vs. 20.8Vpp

```
V2pp = (2.4705 - j*15.52) /

((2.4705-j*15.52)+(15+j*628.2))*V1pp;
```

```
V2pp = abs(V2pp) 0.6426
```
This is close

• $V2pp = 0.6303 Vpp$ using 1st through 4th harmonic to find V2(t)

Summary

Once you have

- Phasors, and
- Superposition

you have Fourier Transforms

Fourier transforms convert any periodic signal into a bunch of sine waves

- Once you have sine waves, you can analyze using phasors and superposition
- The result is more accurate than what we've been doing
- But a *lot* more work

Using the first two terms of the Fourier series (DC $&$ 1st harmonic) will give

- Better answers than what we've been doing, and
- Is just a little harder