Lecture 20: Fourier Transform and Speech Recognition

University of Southern California

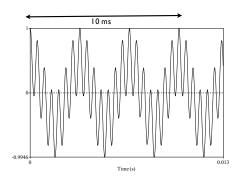
Linguistics 285

USC Linguistics

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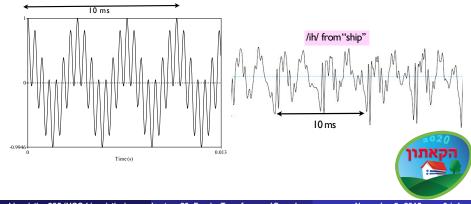
- How do we find the frequencies that compose a signal?
- In a simple case, we can find them by careful observation of a waveform



• But not in a more complex case

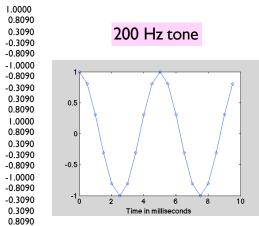


- How do we find the frequencies that compose a signal?
- Observation of waveform in simple, artificial case, but not in complex, real case

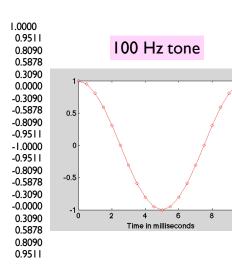


- Fourier Analysis is the technique that does this.
- Produces something called the Fourier Transform, which contains the information in the spectrum.
- But how does it work?
- By using the inner product!
- Take the inner product of the signal (waveform) with pure tones of all possible frequencies.
- The size of the IP tells us how much that signal is similar to each tone.
- That quantitative similarity is the relative amplitude of the frequency in complex signal.

- But how can we take the inner product of a signal with tones?
- Signals are just sequences of numbers, vectors!
- e In this example, a new number (called a sample) every .5 ms









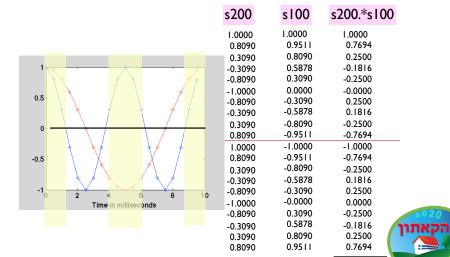
10

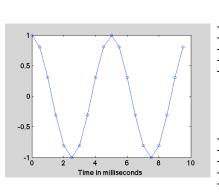
Fourier Analysis via inner product

- Let's treat the 200 Hz tone as our signal whose spectrum we want to measure.
- e Take inner product with 100 Hz, 200 Hz...500 Hz, etc.



There are times where the product is large where both vectors are positive, but each one is exactly canceled by a time when the vectors have opposite signs, so the product is negative. The sum (IP) is zero.

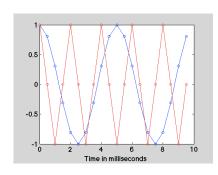




200	s200	s200.*s200
1.0000	1.0000	1.0000
0.8090	0.8090	0.6545
0.3090	0.3090	0.0955
-0.3090	-0.3090	0.0955
-0.8090	-0.8090	0.6545
-1.0000	-1.0000	1.0000
-0.8090	-0.8090	0.6545
-0.3090	-0.3090	0.0955
0.3090	0.3090	0.0955
0.8090	0.8090	0.6545
1.0000	1.0000	1.0000
0.8090	0.8090	0.6545
0.3090	0.3090	0.0955
-0.3090	-0.3090	0.0955
-0.8090	-0.8090	0.6545
-1.0000	-1.0000	1.0000
-0.8090	-0.8090	0.6545
-0.3090	-0.3090	0.0955
0.3090	0.3090	0.0955
0.8090	0.8090	0.6545

Sum = 10





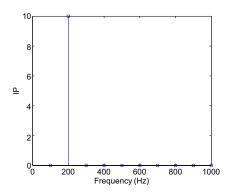
s200	s500	s200.*s500
1.0000	1.0000	1.0000
0.8090	0.0000	0.0000
0.3090	-1.0000	-0.3090
-0.3090	-0.0000	0.0000
-0.8090	1.0000	-0.8090
-1.0000	0.0000	-0.0000
-0.8090	-1.0000	0.8090
-0.3090	-0.0000	0.0000
0.3090	1.0000	0.3090
0.8090	0.0000	0.0000
1.0000	-1.0000	-1.0000
0.8090	-0.0000	-0.0000
0.3090	1.0000	0.3090
-0.3090	-0.0000	0.0000
-0.8090	-1.0000	0.8090
-1.0000	-0.0000	0.0000
-0.8090	1.0000	-0.8090
-0.3090	-0.0000	0.0000
0.3090	-1.0000	-0.3090
0.8090	-0.0000	-0.0000

Sum = 0



Fourier Analysis via inner product

 Other frequencies will also give 0 as the IP, so this our resulting spectrum:



• This will also work with more complex signals!



Example recognizer: Match unknown (test) vowels from one speaker to template vowels from another speaker

- 1. Select a short waveform chunk from each template and test vowel.
- Save waveform chunks to file and import into a matrix in Matlab python
- 3. Derive the **spectrum** of each vowel by using the Fourier Transform: take the inner product of the vowel waveform with the waveforms of a series of tones with different frequencies.
- 4. Compare the **similarity** of each test vowel to each template vowel by taking the inner product of the spectrum of the test vowel with each template vowel:
 - Recognition: The largest IP value obtained for a given test vowel is the vowel that our recognizer selects as the one spoken.

Perform Fourier Transform for each vowel

- Take the inner product of each vowel waveform (vector) with the waveform of tones from 100 Hz to 3000 Hz, in steps of 100 Hz.
- The resulting vector of inner products for each of the 30 frequencies is the spectrum.
- The tone waveforms are stored in a matrix called **freqs**.
- Each column is a different frequency. The rows contain the successive time samples.

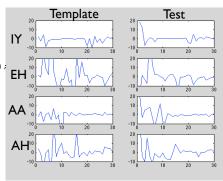
```
>>size (freqs)
ans =
2000 30
```



Perform Fourier Transform for each vowel

Plot the resulting spectra

```
%plot_spec
for i = 1:4
    subplot (4,2,2*i-1), plot(spec_temp(:,i));
    subplot (4,2,2*i-1), ylim([-10 20]);
    subplot (4,2,2*i), plot(spec_test(:,i));
    subplot (4,2,2*i), ylim([-10 20]);
end
```





Compare each unknown vowels to all the templates

- e Go thru each test vowel in a loop.
- For each test vowel, go through each template vowel in a loop and take the IP.
- Save the IP in a matrix in which the rows correspond to the test vowels and the columns the template vowels.

```
for j = 1:4
        IP(j,i) = sum((spec test(:,j)-mean(spec_test(:,j))).*(spec_temp(:,i)-mean(spec_temp(:,i))));
   end
>>IP
TP =
    1.0e + 03 *
             -0.2601
                      0.0546
                                  0.0495
    0.0023
    0.0855
             1.1562
                       0.1832
                                  -0.5468
    -0.0204
             -0.4350
                       0 2976
                                 -0 3989
```



-1 5917

0.0845

0 7764

Results

- How do we interpret the IP matrix?
- Each row has the results for one test vowel, comparing it to all template vowels, across the columns

Template

	IY	EH	AA	AH
	0.00	23 -0.2601	0.0546	0.0495
st	₽ 0.08	55 1.1562	0.1832	-0.5468
Test	AA -0.02	04 -0.4350	0.2976	-0.3989
•	AH -0.064	48 -1.5917	0.0845	0.7764

- The largest IP for each test vowel is the vowel category our system has recognized for that vowel
- If our recognizer worked, the diagonal cells of the matrix should be larger than the off-diagonal cells.
- e 3 vowels were correctly recognized. One was not (/IY/).

Important functions in numpy

- Numpy is a math library in python that does a lot of things
- e linspace: Return evenly spaced numbers over a specified interval
- sin: Computes sinus of number



Important functions in scipy

- Numpy is a science library in python that does a lot of things
- io.wavfile: Reads wav files returns the sample rate and the data
- fftpack.fft: Computes fft of data vector (fourier transform)

