COMP 546

Lecture 19

Sound 2: frequency analysis

Tues. March 27, 2018

Speed of Sound

Sound travels at about 340 m/s, or 34 cm/ms.

(This depends on temperature and other factors)

Wave equation

$$Pressure = I_{atm} + I(X, Y, Z, t)$$

I(X,Y,Z,t) is not an arbitrary function.

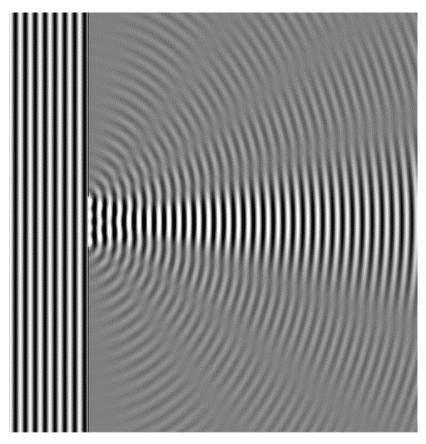
Rather:

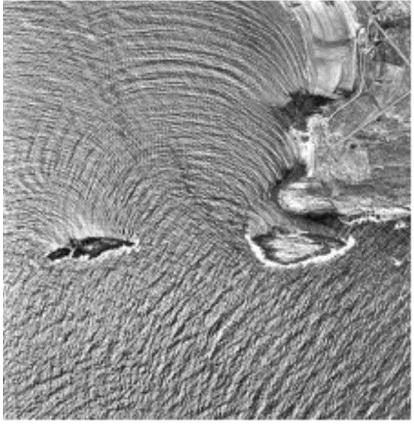
$$\left(\frac{\partial^{2}}{\partial X^{2}} + \frac{\partial^{2}}{\partial Y^{2}} + \frac{\partial^{2}}{\partial Z^{2}}\right) I(X, Y, Z, t) = \frac{1}{v^{2}} \frac{\partial^{2}}{\partial t^{2}} I(X, Y, Z, t)$$

$$v = 340 \text{ m/s}$$

The wave equation + boundary conditions give complicated shadow and reflection effects.

What happens when sound enters the ear?





plane wave + single slit

sea waves + islands

Musical sounds

(brief introduction)

Example: guitar



Write one string displacement at t = 0 as sum of sines.

$$= \left(\begin{array}{c} + \\ + \\ \end{array} \right) + \left(\begin{array}{c} + \\ + \\ \end{array} \right) + \begin{array}{c} + \\ + \end{array} \right) + \begin{array}{c} + \\ + \end{array} \right)$$

Modes are $\sin(\frac{\pi}{L}jx)$ where L is the length of the string, j is an integer.

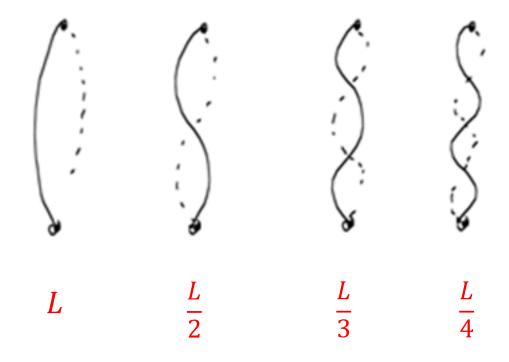
$$L = \left(\begin{array}{c} + \\ + \\ \end{array} \right) + \left(\begin{array}{c} + \\ \end{array} \right) + e^{\dagger}c$$

Physics says:

$$\omega = \frac{C}{L}$$

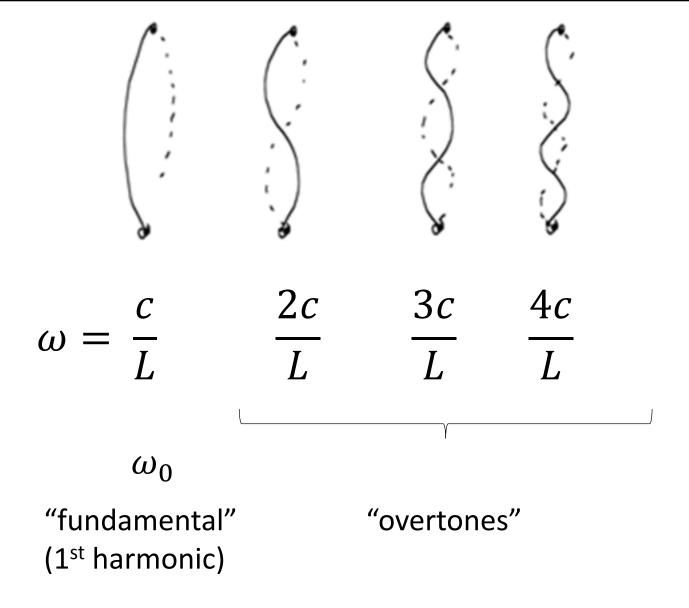
where constant c depends on physical properties of string (mass density, tension)

Modes of a vibrating string each have fixed points which reduce the effective length.



Physics says:

$$\omega = \frac{c}{L} \qquad \frac{2c}{L} \qquad \frac{3c}{L} \qquad \frac{4c}{L}$$



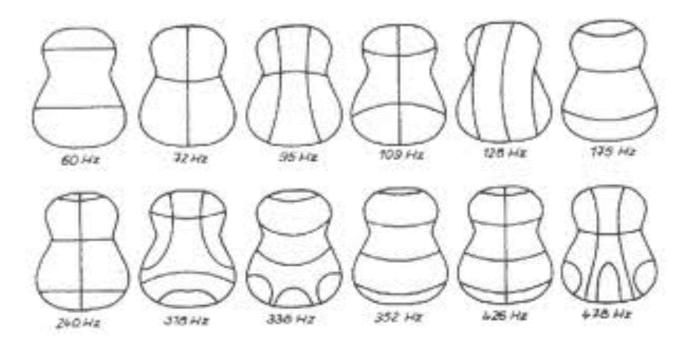
The temporal frequency $\ m\ \omega_0$ is called the m-th harmonic.



For stringed instruments, most of the sound is produced by vibrations of the instrument body (neck, front and back plates).

http://www.acs.psu.edu/drussell/guitars/hummingbird.html

The lines in the sketches below are the nodal points. They don't move.



These are vibration *modes*, not harmonics. The guitar sound is a sum of these modes.

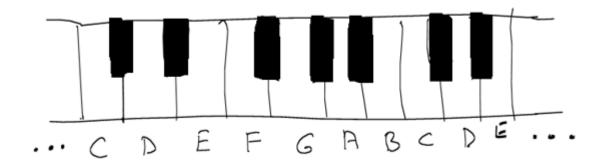
Difference of two frequencies ω_1 and ω_2 :

$$log_2 \frac{\omega_2}{\omega_1}$$
 octaves.

e.g. 1 octave is a doubling of frequency.

(Western) Musical Notes

Each "octave" ABCDEFGA is divided into 12 "semitones", separated into 1/12 octave.



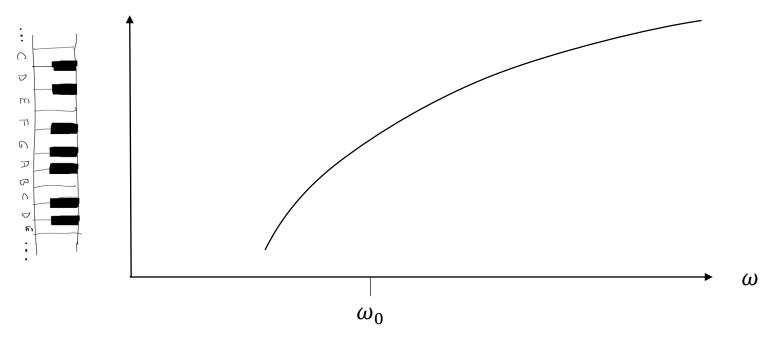
C-D, D-E, F-G, G-A, A-B are two semitones each E-F, B-C are one semitone each.

Q: How many semi-tones are there from ω_0 to ω ?

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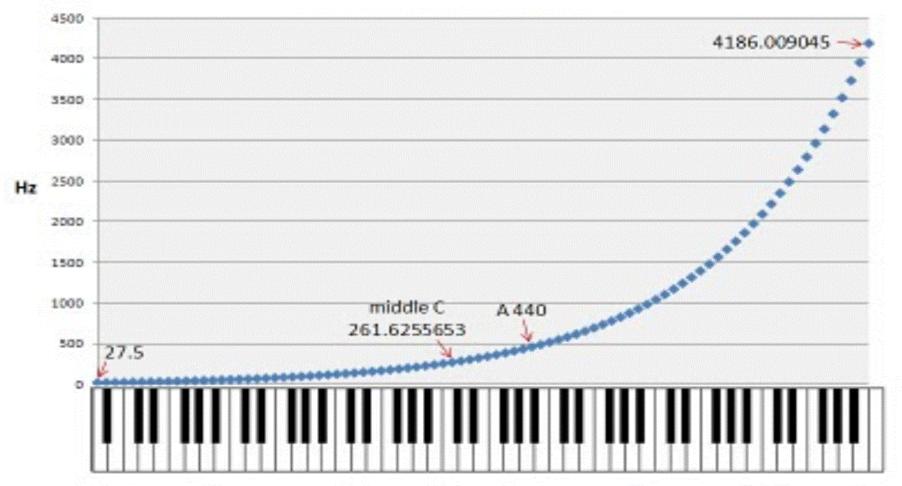
A:

12
$$log_2 \frac{\omega}{\omega_0}$$



Fundamental frequency of note

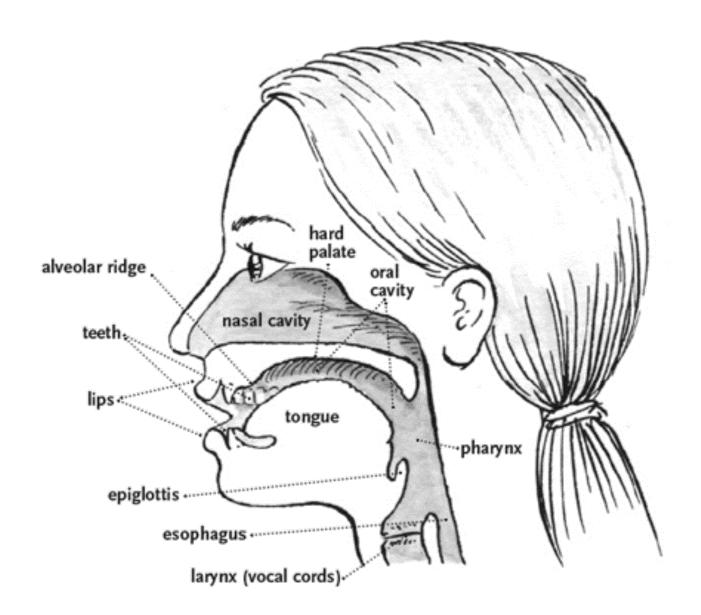
88 fundamental frequencies (Hz) on a keyboard



The fundamental frequencies of successive notes define a geometric progression.

This is different from the harmonics of a vibrating string which define an arithmetic progression.

Speech Sounds



What determines speech sounds?

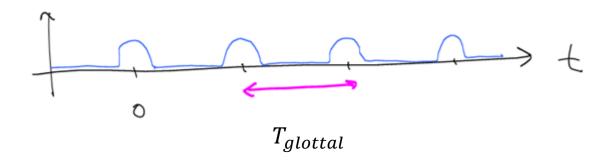
voiced vs. unvoiced

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'zzzz' vs. 'ssss', 'vvvv' vs. 'ffff'
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articulators (jaw, tongue, lips)

'aaaa', 'eeee', 'oooo', ...

Voiced sounds are produced by "glottal pulses".

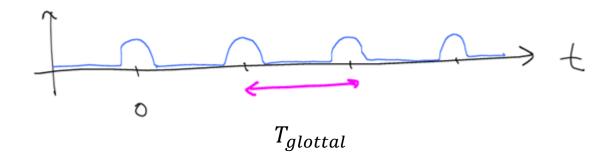


$$\sum_{j=0}^{n_{glottal}} g(t-j T_{glottal})$$

Exercise 16 Q7.

$$g(t - t_0) = g(t) * \delta(t - t_0)$$

Voiced sounds are produced by "glottal pulses".



$$\sum_{j=0}^{n_{glottal}} g(t - j T_{glottal}) = g(t) * \sum_{j=0}^{n_{glottal}} \delta(t - j T_{glottal})$$

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$$T_{glottal}$$

$$\sum_{j=0}^{n_{glottal}} g(t-j T_{glottal})$$

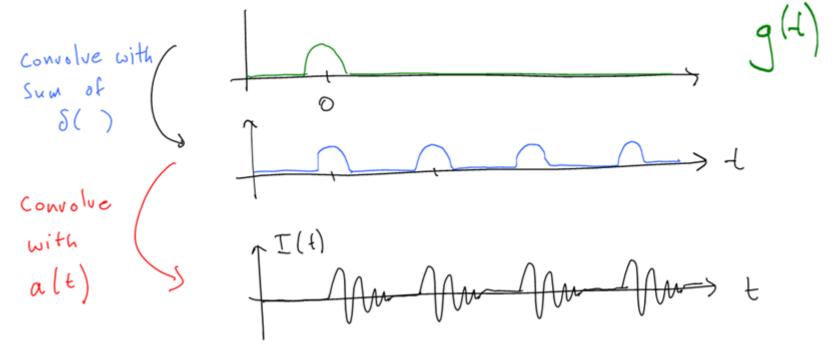
decrease $T_{glottal}$ by increasing tension in vocal cords

increase frequency of pulses

Let a(t) be the impulse response function of the articulators.

(jaw, tongue, lips)

$$I(t) = a(t) * g(t) * \sum_{j=0}^{n_{glottal}} \delta(t - j T_{glottal})$$



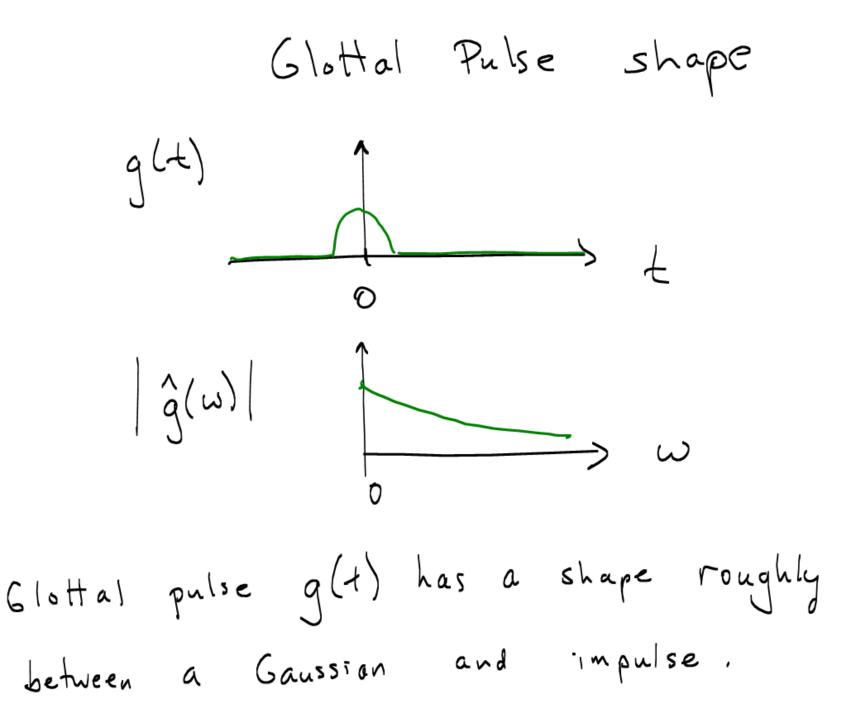
Q: What is the Fourier transform of $T(t) = \alpha(t) * g(t) * \sum_{j=0}^{n_{pulse}-1} \delta(t-jt) ?$ T (1)

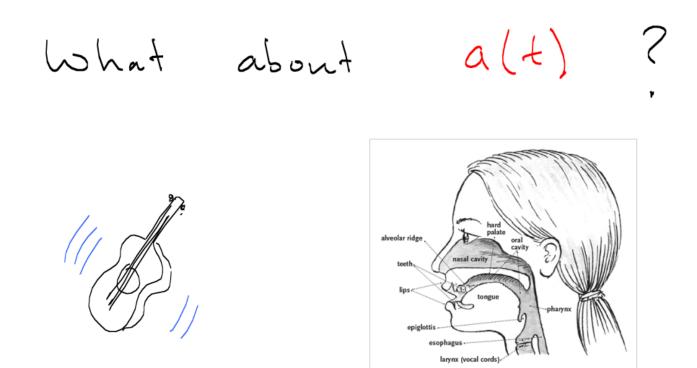
Am Am Am > t

Q: What is the Fourier transform of
$$T(t) = a(t) * g(t) * \int_{j=0}^{n_{pulse}-1} f(t) dt$$

$$T(t) = a(t) * g(t) * \int_{j=0}^{n_{pulse}-1} f(t) dt$$

$$T(\omega) = \hat{a}(\omega) \cdot \hat{g}(\omega) \cdot F \int_{j=0}^{n_{pulse}-1} f(t) dt$$

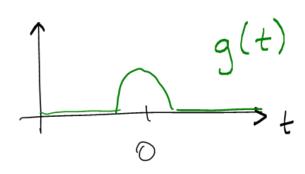


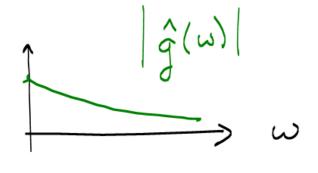


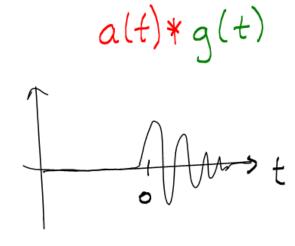
Oral and nasal cavity have resonant modes of vibration, like air cavity in guitar does.

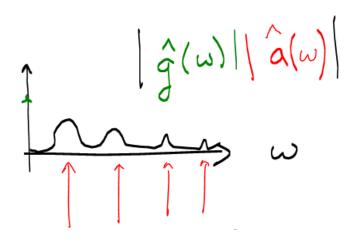
Time domain

Temporal frequency domain









Peaks are called "formants"

$$\mathbf{F} \sum_{j=0}^{n_{glottal}} \delta(t - j T_{glottal}) = ?$$

 T_g is the period of the glottal pulse train.

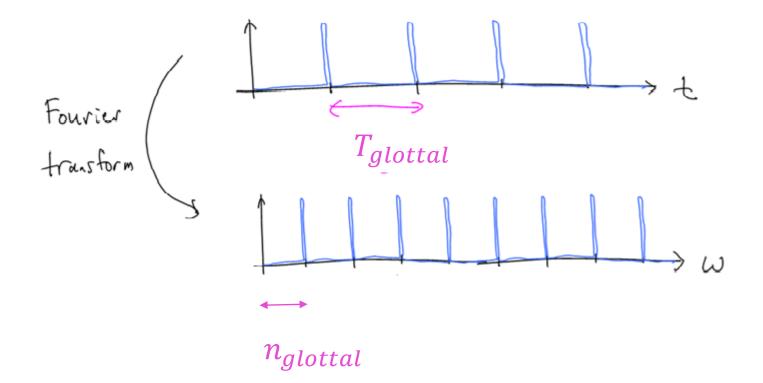
The pulse train has $n_{glottal}$ pulses in T time steps, i.e.

$$T_{glottal} n_{glottal} = T.$$

Assume that the Fourier transform is taken over T samples.

Assignment 3: Show

$$\mathbf{F} \sum_{j=0}^{n_{glottal}-1} \delta(t-j T_{glottal}) = n_{glottal} \sum_{m=0}^{T_{glottal}-1} \delta(\omega - m n_{glottal})$$



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Units of temporal frequency ω

 $T_{glottal}$ is the period of the glottal pulse train. $n_{glottal}$ pulses in T time samples.

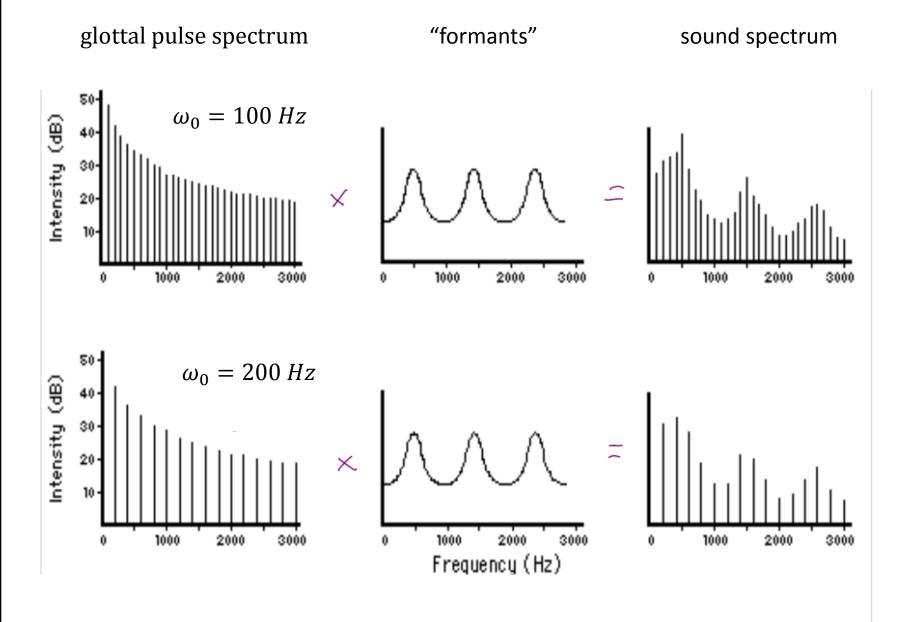
To convert $n_{glottal}$ to 'pulses per second', we divide T (to get pulses per sample) and then multiply by 'time samples per second'. High quality audio uses 44,100 samples per second.

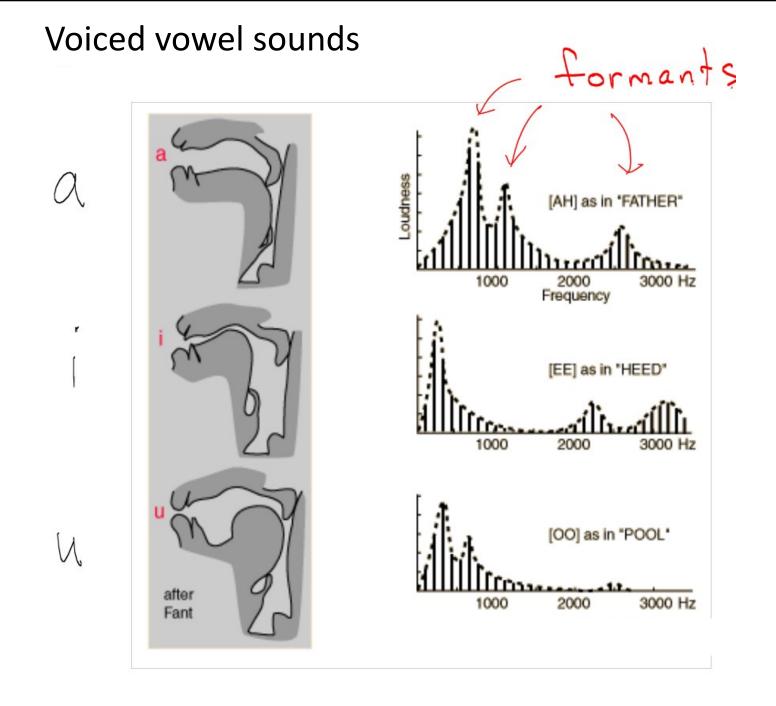
 $n_{glottal}$ is the fundamental frequency of the voiced sound. It determines the "pitch".

Adult males: 100-150

Adult females: 150-250 Hz

Children: over 250 Hz





Unvoiced sounds noise instead of glottal pulses

$$T(t) = a(t) * n(t)$$

Unvoiced sounds noise instead of glottal pulses

$$T(t) = a(t) * n(t)$$

$$\hat{T}(\omega) = \hat{a}(\omega) \hat{n}(\omega)$$

Flat amplitude spectrum on average ('white noise')

Consonants

Restrict flow of air by moving tongue, lips into contact with the teeth & palate.

Fricatives

- voiced z, v, zh, th (the)
- unvoiced ?

Stops

- voiced b, d, g
- unvoiced ?

Nasals (closed mouth)

- m, n, ng

Consonants

Restrict flow of air by moving tongue, lips into contact with the teeth & palate.

Fricatives

- voiced z, v, zh, th (the)
- unvoiced s, f, sh, th (theta)

Stops

- voiced b, d, g
- unvoiced p, t, k

Nasals (closed mouth)

- m, n, ng

I did not have time to cover the following slides properly.

I will present them again in lecture 22.

Spectrogram

Partition a sound signal into B blocks of T samples each (i.e. the sound has BT samples in total).

Take the Fourier transform of each block.

Spectrogram

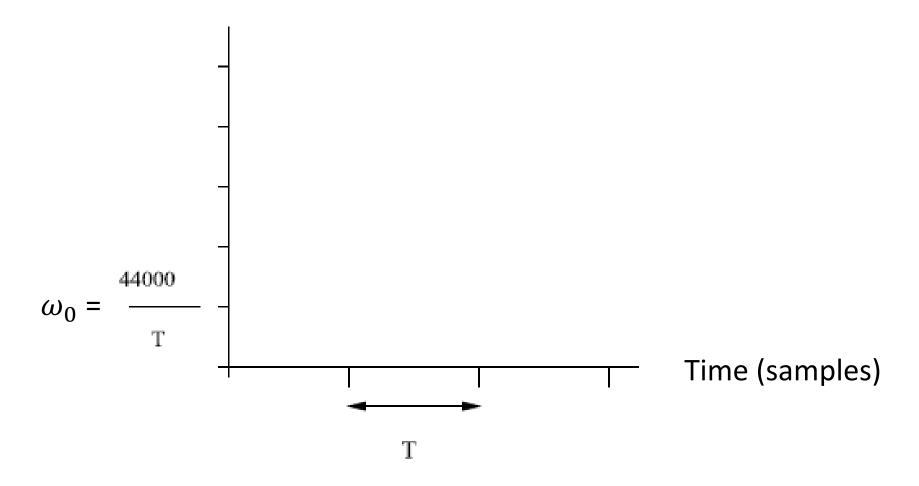
Partition a sound signal into B blocks of T samples each (i.e. the sound has BT samples in total).

Take the Fourier transform of each block.

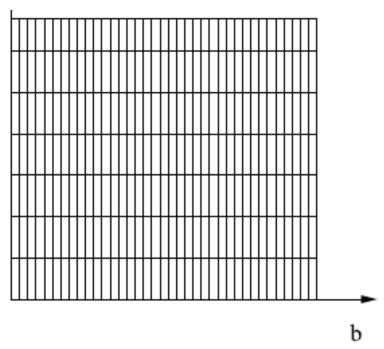
Let b be the block number, and ω units be cycles per block.

$$\hat{I}(b,\omega) = \sum_{t=0}^{T-1} I(b T + t) e^{-i\frac{2\pi}{T}\omega t}$$

Cycles per second (Hz)

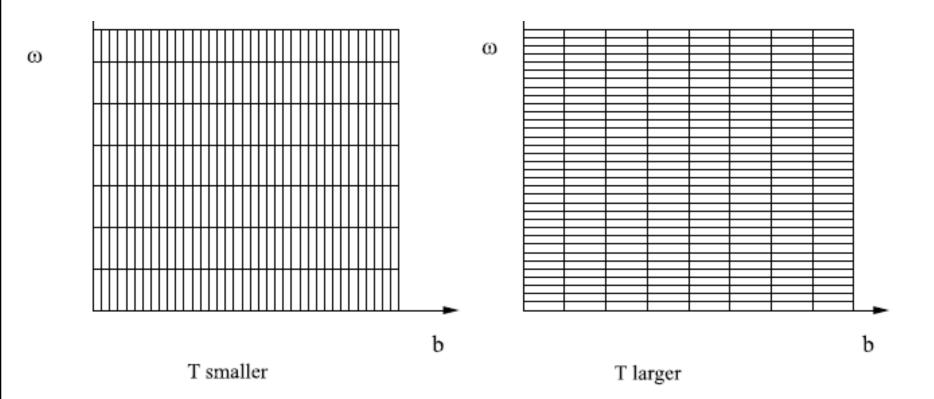






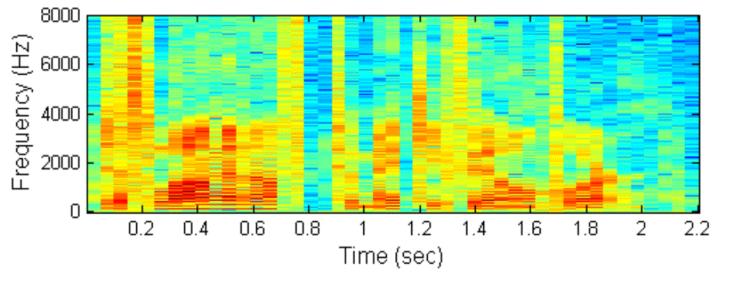
T smaller

e.g. T = 512 samples (12 ms),
$$\omega_0$$
 = 86 Hz



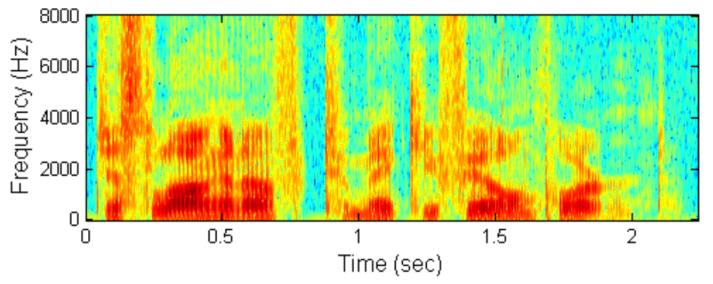
e.g. T = 512 samples (12 ms), ω_0 = 86 Hz T = 2048 samples (48 ms), ω_0 = 21 Hz

You cannot simultaneously localize the frequency and the time. This is a fundamental tradeoff. We have seen it before (recall the Gaussian).



Narrowband

(good frequency resolution, poor temporal resolution ... ~50ms)



Wideband

(poor frequency resolution, good temporal resolution)

Examples: Spectrograms of 10 vowel sounds

