

Lecture 9: Exam 1 Review

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ECE 537: Speech Processing Fundamentals

- 1 Administrative Details
- 2 Loudness
- 3 Vocoder
- 4 Pitch
- 5 Acoustics of Nasal Consonants
- 6 Conclusion

Outline

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Exam 1: Administrative Details

- In class, Wednesday; if you need conflict exam or on-line exam, contact me in advance
- One page handwritten notes, both sides
- No calculator

Content

- Loudness: Intensity, Loudness Level, Masking
- Vocoder: Voiced, Unvoiced, Spectral shape
- Pitch: Autocorrelation, Narrowband signals
- Nasals: Laplace Transform, Plane Waves, Susceptance

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Solutions to the Wave Equation

$$-\frac{\partial^2 p}{\partial x^2} = \frac{1}{c^2} \frac{\partial^2 p}{\partial t^2}$$

The solution to the 1d wave equation is any combination of a rightward-traveling wave, $r(t)$ and a leftward-traveling wave, $l(t)$:

$$p(x, t) = r\left(t - \frac{x}{c}\right) + l\left(t + \frac{x}{c}\right)$$
$$v(x, t) = \frac{1}{\rho c} \left(r\left(t - \frac{x}{c}\right) - l\left(t + \frac{x}{c}\right) \right)$$

Acoustic Intensity of a Pure Tone

Suppose that $p(t)$ is a pure tone, with a root-mean-squared (RMS) amplitude of P Pascals, and a frequency of f Hertz.

$$p(t) = \sqrt{2}P \cos(2\pi ft)$$
$$v(t) = \frac{\sqrt{2}P}{\rho c} \cos(2\pi ft)$$

The intensity of this wave is:

$$J = \langle pv \rangle = f \int_0^{1/f} p(t)v(t) dt$$
$$= f \int_0^{1/f} \frac{2P^2}{\rho c} \cos^2(2\pi ft) dt = \frac{P^2}{\rho c}$$

Sound Pressure Level

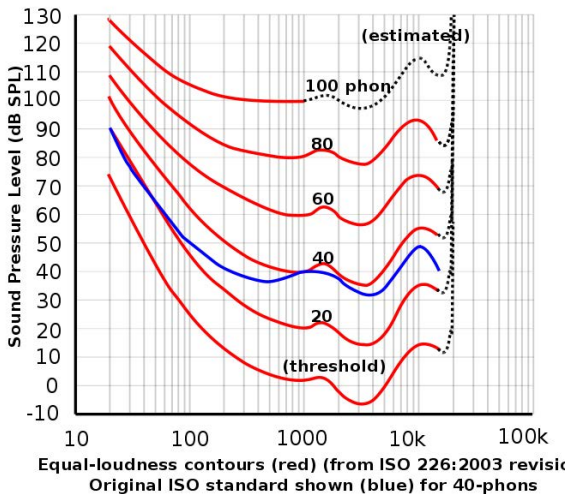
The intensity level of a sound can be measured with respect to a standard reference level. The standard reference level is $J_r = 10^{-12}$ Watts per square meter.

The level of a sound, measured w.r.t. 10^{-12} W/m², is called its “sound pressure level” (SPL). So

$$\beta = 10 \log_{10} \left(\frac{J}{J_r} \right)$$

... has units of “dB SPL.”

Loudness Level



Loudness

$$G(L) = \sum_k b_k G(L_k)$$

$G(L_k)$ is a nonlinear function of the loudness level, L_k . The exam will give you a table of these values.

If you want to find the loudness level, L , of the whole sound, you can use

$$L = G^{-1} \left(\sum_k b_k G(L_k) \right)$$

Masking

- If $\Delta f = |f_2 - f_1| < B$, then just add the intensities of the two tones, and calculate loudness from that.
($B \in \{100, 200, 400, 800\}$, depending on f_2).
- If $\Delta f \geq B$, then

$$b_2 = \left[\frac{250 + \Delta f}{1000} \right] Q(L_2)$$

where $Q(L_2)$ is a nonlinear function of L_2 . The exam will give you a table of its values.

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Voiced Source: Impulse Train

$$\begin{aligned}x[n] &= \sum_{m=-\infty}^{\infty} \delta[n - mN] \\ &= \frac{1}{N} \sum_{k=0}^{N-1} e^{j\frac{2\pi kn}{N}}\end{aligned}$$

Spectrum of a Bandpass-Filtered Impulse Train

Suppose $x[n]$ is periodic:

$$x[n] = \sum_{k=0}^{N-1} X_k e^{j\frac{2\pi kn}{N}},$$

and we bandpass filter it with a filter $h[n]$:

$$y[n] = h[n] * x[n],$$

then $y[n]$ is periodic with Fourier series coefficients given by:

$$Y_k = H\left(\frac{2\pi k}{N}\right) X_k$$

Unvoiced Source: White Noise

The autocorrelation of a wide-sense stationary signal is:

$$R_{xx}[m] = E [x[n]x[n + m]]$$

Its power spectrum is:

$$R_{xx}(\omega) = E \left[\frac{1}{N} |X(\omega)|^2 \right] = \mathcal{F} \{ R_{xx}[m] \}$$

A unit variance white noise signal has

$$R_{xx}[m] = \delta[m]$$

$$R_{xx}(\omega) = 1$$

Spectrum of a Bandpass-Filtered Noise

$$y[n] = h[n] * x[n]$$

$$R_{yy}[n] = h[n] * h^*[-n] * R_{xx}[n]$$

$$R_{yy}(\omega) = |H(\omega)|^2 R_{xx}(\omega)$$

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Correlogram

- 1 Pass the signal through a bank of bandpass filters:

$$x_f[n] = h_f[n] * x[n],$$

where f denotes the center frequency, in Hertz, and we assume that the bandwidth is one auditory critical band.

- 2 Compute the autocorrelation in each channel:

$$\phi(f, m) = E [x_f[n]x_f[n + m]]$$

Autocorrelation of a Sinusoid

Suppose $x[n]$ is periodic, and the critical band contains only one harmonic:

$$x_f[n] = A \cos\left(\frac{2\pi kF_0}{F_s} n + \theta\right)$$

where kF_0 is within the passband of the filter centered at f . Suppose we treat the timing, n , as a random variable. Then

$$\begin{aligned}\phi(f, m) &= E_n [x_f[n]x_f[n + m]] \\ &= \frac{A^2}{2} \cos\left(\frac{2\pi kF_0}{F_s} m\right)\end{aligned}$$

... which is periodic with a period of $\frac{1}{kF_0}$, and at every multiple thereof, including the pitch period.

Autocorrelation of two sinusoids

Suppose $x[n]$ is periodic, and the critical band contains only two harmonics:

$$x_f[n] = A_k \cos\left(\frac{2\pi k F_0}{F_s} n + \theta_k\right) + A_{k+1} \cos\left(\frac{2\pi(k+1)F_0}{F_s} n + \theta_{k+1}\right)$$

where kF_0 and $(k+1)F_0$ are within the passband of the filter centered at f .

Suppose we treat the timing, n , as a random variable. Then

$$\begin{aligned}\phi(f, m) &= E_n [x_f[n]x_f[n+m]] \\ &= \frac{A_k^2}{2} \cos\left(\frac{2\pi k F_0}{F_s} m\right) + \frac{A_{k+1}^2}{2} \cos\left(\frac{2\pi(k+1)F_0}{F_s} m\right)\end{aligned}$$

... which is periodic at the pitch period $\frac{1}{F_0}$.

Autocorrelation of Narrowband Noise

Suppose $x[n]$ is unit-variance white noise. Then

$$\phi(f, m) = E_n [x_f[n]x_f[n + m]] = h_f[m] * h_f^*[-m] * R_{xx}[m]$$

But what is that? It turns out to be easier to solve in the frequency domain:

$$\begin{aligned}\phi(f, \omega) &= |H_f(\omega)|^2 R_{xx}(\omega) = |H_f(\omega)|^2 \\ &= \begin{cases} 1 & \frac{2\pi(f-B/2)}{F_s} \leq |\omega| \leq \frac{2\pi(f+B/2)}{F_s} \\ 0 & \text{otherwise} \end{cases}\end{aligned}$$

where B is the auditory filter bandwidth. This has the inverse transform of

$$\phi(f, m) = \left(\frac{B}{F_s}\right) \operatorname{sinc}\left(\frac{\pi B}{F_s} m\right) \cos\left(\frac{2\pi f}{F_s} m\right)$$

... which is periodic with a period of $\frac{1}{f}$, which varies from filter to filter, and has no relationship to any overall pitch period.

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The Two-Sided Laplace Transform

$$X(s) = \int_{-\infty}^{\infty} x(t)e^{-st} dt$$

Example:

$$x(t) = e^{at} u(t)$$

$$X(s) = \frac{1}{s - a}, \quad \text{for } \Re(s) > \Re(a)$$

Delay Property

If

$$y(t) = x(t - d),$$

then

$$\begin{aligned} Y(s) &= \int_{-\infty}^{\infty} y(t) e^{-st} dt \\ &= X(s) e^{-sd} \end{aligned}$$

Laplace Transform of the Solution to the Wave Equation

$$p(x, t) = r\left(t - \frac{x}{c}\right) + l\left(t + \frac{x}{c}\right)$$

Let's take the Laplace transform of that:

$$\begin{aligned} P(x, s) &= \int_{-\infty}^{\infty} p(x, t) e^{-st} dt \\ &= R(s) e^{-xs/c} + L(s) e^{xs/c} \end{aligned}$$

Volume Velocity

The relationship between pressure and volume velocity is:

$$P(x, s) = R(s)e^{-sx/c} + L(s)e^{sx/c},$$
$$U(x, s) = \frac{A(x)}{\rho c} \left(R(s)e^{-sx/c} - L(s)e^{sx/c} \right)$$

The Zero-Pressure Constraint at the Lips

$$p(x, t) = r\left(t - \frac{x}{c}\right) + l\left(t + \frac{x}{c}\right),$$
$$P(x, s) = R(s)e^{-sx/c} + L(s)e^{sx/c}.$$

If we apply the condition that $p(d_l, t) = 0$, we learn that $l(t)$ is a reflection of $r(t)$, delayed by $2d_l/c$ and multiplied by -1:

$$B(x, s) = \frac{U(x, s)}{P(x, s)}$$
$$= -\frac{A(x)}{\rho c} \coth(s(x - d_l)/c)$$

The Zero-Velocity Constraint at the Lips

$$p(x, t) = r\left(t - \frac{x}{c}\right) + l\left(t + \frac{x}{c}\right),$$
$$P(x, s) = R(s)e^{-sx/c} + L(s)e^{sx/c}.$$

If we apply the condition that $u(d_l, t) = 0$, we learn that $l(t)$ is a reflection of $r(t)$, delayed by $2d_l/c$ and multiplied by -1:

$$B(x, s) = \frac{U(x, s)}{P(x, s)}$$
$$= -\frac{A(x)}{\rho c} \tanh(s(x - d_l)/c)$$

Resonances of a Nasal Consonant

Fujimura proposed computing the resonances of a nasal consonant by finding the zeros of the total susceptance,

$$B(s) = B_p(s) + B_n(s) + B_m(s)$$

For the consonant /ŋ/, Fujimura assumed that the mouth cavity has zero volume, thus $B_m(s) = 0$, so resonances of /ŋ/ are the zeros of $B_i(s) = B_n(s) + B_p(s)$.

The resonances of /m/ and /n/ are then modeled by the equation

$$B_i(s) = -B_m(s)$$

Anti-resonance

The anti-resonance (the zeros of the transfer function) are the frequencies at which

$$B_m(s) = \infty$$

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Conclusion: Topics for Exam

- Loudness: Intensity, Loudness Level, Masking
- Vocoder: Voiced, Unvoiced, Spectral shape
 - Not covered: Brownian motion, relaxation oscillator
- Pitch: Autocorrelation, Narrowband signals
 - Not covered: Gammatone filters
- Nasals: Laplace Transform, Plane Waves, Susceptance