

Lecture 11

ECE5022: Digital Signal Processing

Lecture 11: Homomorphic Signal Processing

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Announcements

- Term Project
 - Due: 6/14
- Final Exam
 - 4:30PM – 5:30PM (60 Minutes) on Tuesday, 6/7
 - Place: Hi-Tech Center Room 228
- Grading
 - Exams: 30% (Midterm) + 30% (Final)
 - Project: 20%, Attendance: 10%, Homework: 10%

open book No computer / tablet / phone.

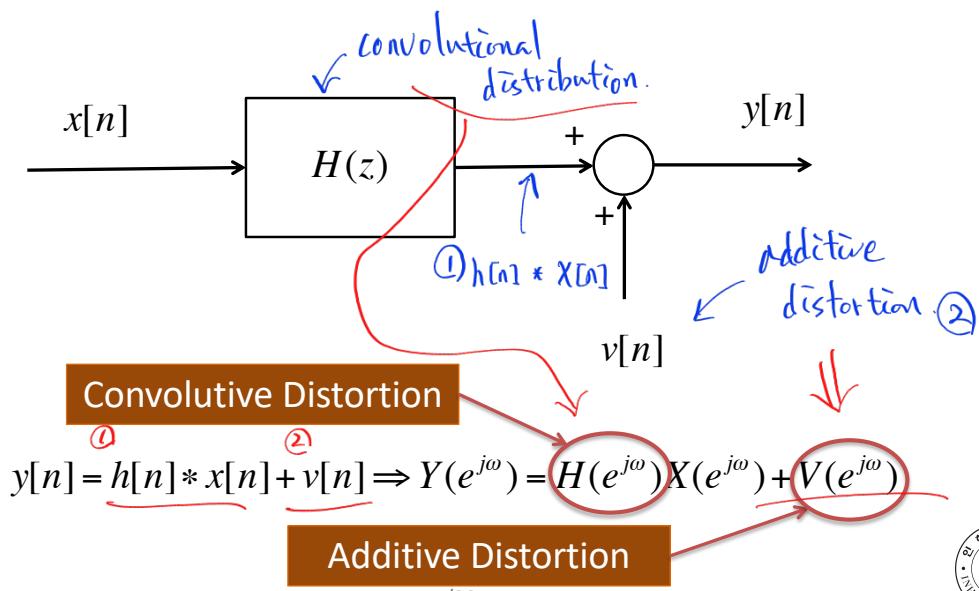
(FFT 33%)
중간고사 전형성.)

Outline

- Homomorphic Signal Processing
- Cepstrum Definitions *(Analysis)*
 - Real Cepstrum
 - Complex Cepstrum
- Homomorphic Deconvolution
- Speech Processing Applications



Signal Distortion Model



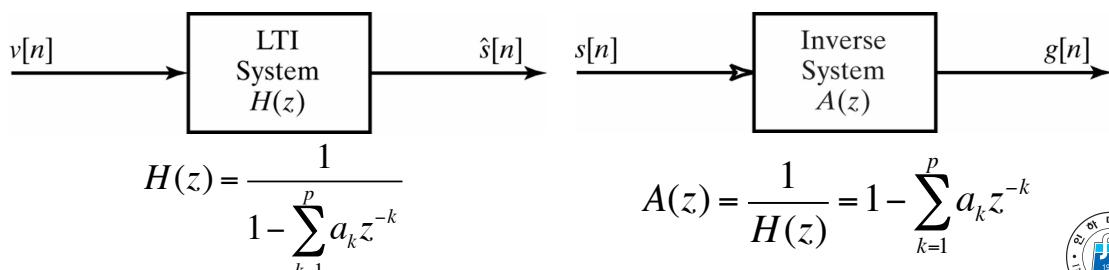
Two Types of Signal Distortion

- Additive Noise
 - Distortion caused by a noise signal **added** to the input
 - Example: Wiener Filtering for reducing additive noise
- Convulsive Noise
 - Distortion caused by **convolving (filtering)** the input with a noise signal



Removing Convulsive Noise

- Convulsive Noise
 - Need to find an inverse filter to remove it
 - Example: All-pole modeling gives an FIR inverse filter



Homomorphic Signal Processing

- Parametric signal modeling provides a simple inverse filter as far as the model is good enough
- Nonparametric approach may be required if the model is not good enough
problem to solve.
- Convert the problem to an easier ~~solution~~
— Transform convolutive noise to additive noise



Superposition Property

- Given a linear operator $L[\bullet]$

$$L[a_1x_1[n] + a_2x_2[n]] = a_1L[x_1[n]] + a_2L[x_2[n]]$$

- Equivalently

\uparrow
z 변환

$$L[a_1X_1(z) + a_2X_2(z)] = a_1L[X_1(z)] + a_2L[X_2(z)]$$

- Wiener filter can be used to remove the additive noise



Generalized Superposition

- Given a nonlinear operator Φ that satisfies

$$\Phi[X_1(z)X_2(z)] = \Phi[X_1(z)] + \Phi[X_2(z)]$$

- If such a nonlinear operation exists, we can treat the convolutive noise as additive noise
- Homomorphic signal processing
 - Convert the problem to an easier problem and solve
 - Transform the output signal back to the original domain



Complex Logarithm I/II

- We can easily see that

$$\log_e[X_1(z)X_2(z)] = \log_e[X_1(z)] + \log_e[X_2(z)]$$

- For the DTFT (complex valued)

$$\begin{aligned} \log_e[X(e^{j\omega})] &= \log_e[|X(e^{j\omega})| e^{j\angle X(e^{j\omega})}] \\ &= \log_e[|X(e^{j\omega})|] + \log_e[e^{j\angle X(e^{j\omega})}] \\ &= \log_e[|X(e^{j\omega})|] + j\angle X(e^{j\omega}) \end{aligned}$$

$$X(e^{j\omega}) = |X(e^{j\omega})| e^{j\angle X(e^{j\omega})}$$



deg of the magnitude of 5 times phase?

Complex Logarithm II/II

- Suppose

$$X(e^{j\omega}) = X_1(e^{j\omega})X_2(e^{j\omega})$$

- Then

$$\begin{aligned} \log[X(e^{j\omega})] &= \log \left[\left(|X_1(e^{j\omega})| e^{j\angle X_1(e^{j\omega})} \right) \left(|X_2(e^{j\omega})| e^{j\angle X_2(e^{j\omega})} \right) \right] \\ &= \log \left[|X_1(e^{j\omega})| |X_2(e^{j\omega})| e^{j(\angle X_1(e^{j\omega}) + \angle X_2(e^{j\omega}))} \right] \\ &= \log |X_1(e^{j\omega})| + \log |X_2(e^{j\omega})| + j(\angle X_1(e^{j\omega}) + \angle X_2(e^{j\omega})) \end{aligned}$$

↑
even odd magnitude phase



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기프트럼



Real Cepstrum

- For a real sequence $x[n]$
 - $|X(e^{j\omega})|$ is even and $\angle X(e^{j\omega})$ is odd and both are real
 - $\log |X(e^{j\omega})|$ is real and even and $j\angle X(e^{j\omega})$ is imaginary and odd
- If we define

$$C_x(e^{j\omega}) = \log |X(e^{j\omega})|$$

whose inverse DTFT is defined as the Real Cepstrum

$$\underline{c_x[n]} = \frac{1}{2\pi} \int_{-\pi}^{\pi} C_x(e^{j\omega}) e^{j\omega n} d\omega = \frac{1}{2\pi} \int_{-\pi}^{\pi} \log |X(e^{j\omega})| e^{j\omega n} d\omega$$

Homomorphic Signal Processing

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Real Cepstrum

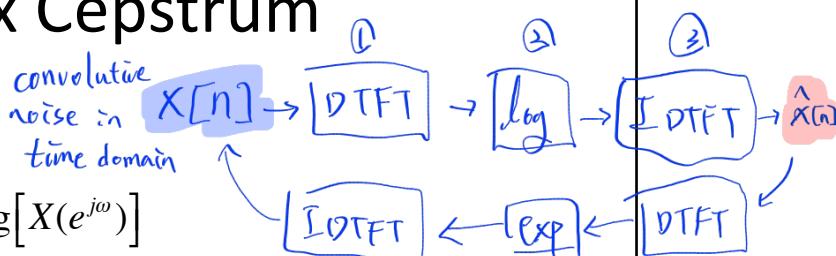
1549ms!

Complex Cepstrum

- If we define

$$\hat{X}(e^{j\omega}) = \log [X(e^{j\omega})]$$

- The complex cepstrum is defined as



is real

inverse
dft

$$\hat{x}[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} \hat{X}(e^{j\omega}) e^{j\omega n} d\omega = \frac{1}{2\pi} \int_{-\pi}^{\pi} \log [X(e^{j\omega})] e^{j\omega n} d\omega$$

$$= \frac{1}{2\pi} \int_{-\pi}^{\pi} [\log |X(e^{j\omega})| + j\angle X(e^{j\omega})] e^{j\omega n} d\omega$$

Even

real part

Odd

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Invertibility of Complex Cepstrum

- The inverse DTFT is invertible, i.e., DTFT
- The inverse of logarithm is invertible, i.e., exponential
- The complex cepstrum is therefore invertible
- The cepstrum of $X(e^{j\omega}) = X_1(e^{j\omega})X_2(e^{j\omega})$

$$\begin{aligned}\hat{X}(e^{j\omega}) &= \hat{X}_1(e^{j\omega}) + \hat{X}_2(e^{j\omega}) \\ &= \log|X(e^{j\omega})| + j\angle X(e^{j\omega})\end{aligned}$$

- The cepstrum is guaranteed to be real for real $x[n]$



Some Terminologies

- Quefrency
 - The time index n for the cepstrum

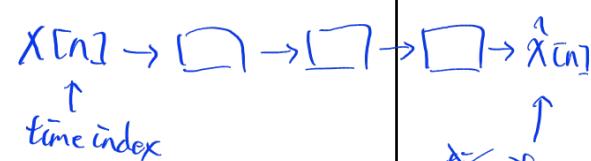
$$\hat{x}[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} \log[X(e^{j\omega})] e^{j\omega n} d\omega$$

filtering

– Reverse of frequency \rightarrow quefrency

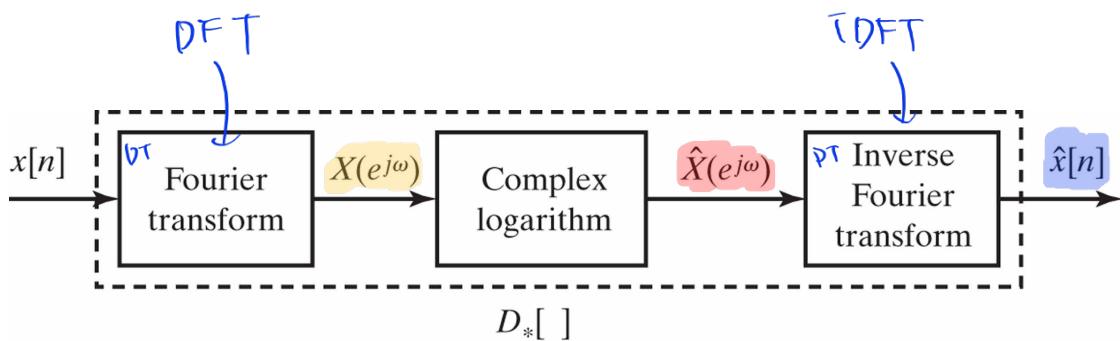
- Liftering

- Filtering in the cepstrum domain
- Reverse of filtering \rightarrow liftering



At the bottom right, there is a circular logo of KAIST (Korea Advanced Institute of Science and Technology) with the year 1954. To its right, the text "example" is written vertically, followed by "note!!".

Complex Cepstrum Diagram



Computation of Complex Cepstrum

- The computation is represented as

$$X(e^{j\omega}) = \sum_{n=-\infty}^{\infty} x[n] e^{-j\omega n}$$

DTFT

$$\hat{X}(e^{j\omega}) = \log[X(e^{j\omega})]$$

Complex Logarithm

$$\hat{x}[n] = \frac{1}{2\pi} \int_{-\pi}^{\pi} \hat{X}(e^{j\omega}) e^{j\omega n} d\omega$$

Inverse DTFT



Complex Cepstrum in DFT

- Sampling the DTFT

$$X[k] = X(e^{j\omega}) \Big|_{\omega=\frac{2\pi k}{N}} = \sum_{n=-\infty}^{\infty} x[n] e^{-j\frac{2\pi k}{N} n}$$

DFT

FFT

$$\hat{X}[k] = \log \left[X(e^{j\omega}) \right] \Big|_{\omega=\frac{2\pi k}{N}}$$

Complex Logarithm

$$\hat{x}_p[n] = \frac{1}{N} \sum_{k=0}^{N-1} \hat{X}[k] e^{j\frac{2\pi n}{N} k}$$

Inverse DFT

IFFT

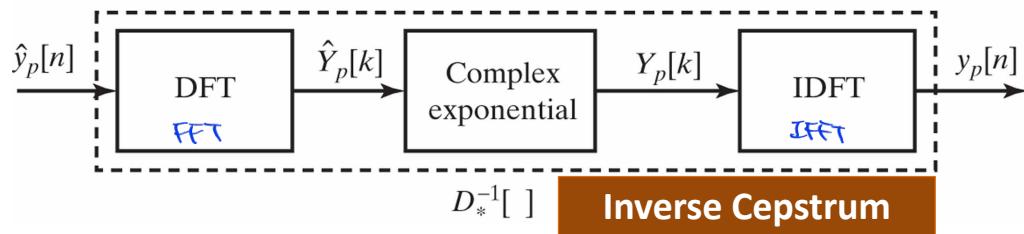
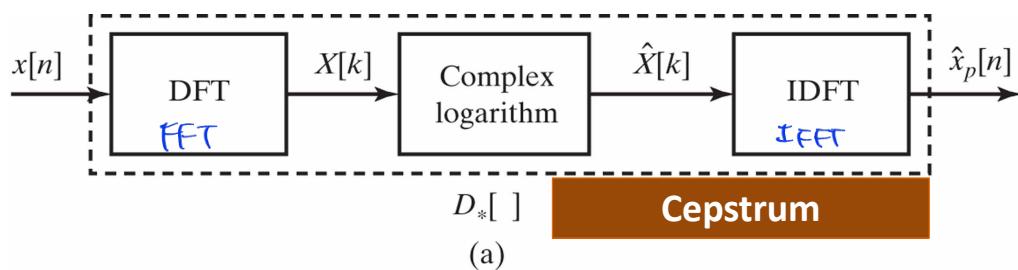


유사해서 쓰인다.

정답과는
다른,
다른.

incorrect!!

Cepstrum and Inverse Cepstrum in DFT

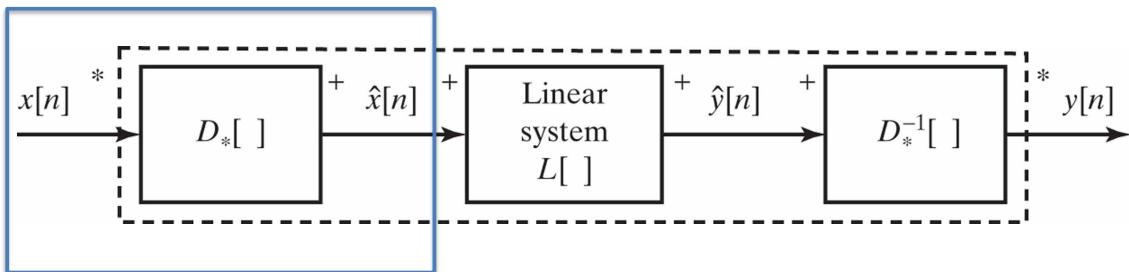


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Homomorphic Signal Processing I/II



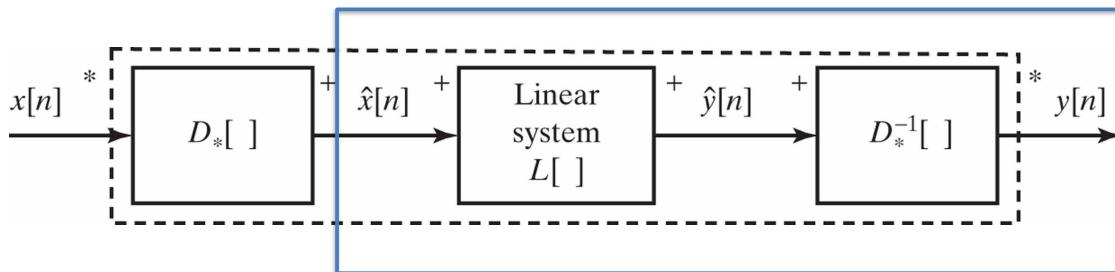
$$x[n] = x_1[n] * x_2[n] \Rightarrow X(z) = X_1(z) \cdot X_2(z)$$

$$\hat{X}(z) = \log[X(z)] = \log[X_1(z)] + \log[X_2(z)] = \hat{X}_1(z) + \hat{X}_2(z)$$

$$\hat{x}[n] = D_*[x_1[n] * x_2[n]] = \hat{x}_1[n] + \hat{x}_2[n]$$



Homomorphic Signal Processing II/II



$$\hat{x}[n] = \hat{x}_1[n] + \hat{x}_2[n]$$

$$\hat{y}[n] = L[\hat{x}_1[n] + \hat{x}_2[n]] = L[\hat{x}_1[n]] + L[\hat{x}_2[n]] = \hat{y}_1[n] + \hat{y}_2[n]$$

$$\hat{y}[n] = D_*^{-1}[\hat{y}_1[n] + \hat{y}_2[n]] = y_1[n] * y_2[n]$$

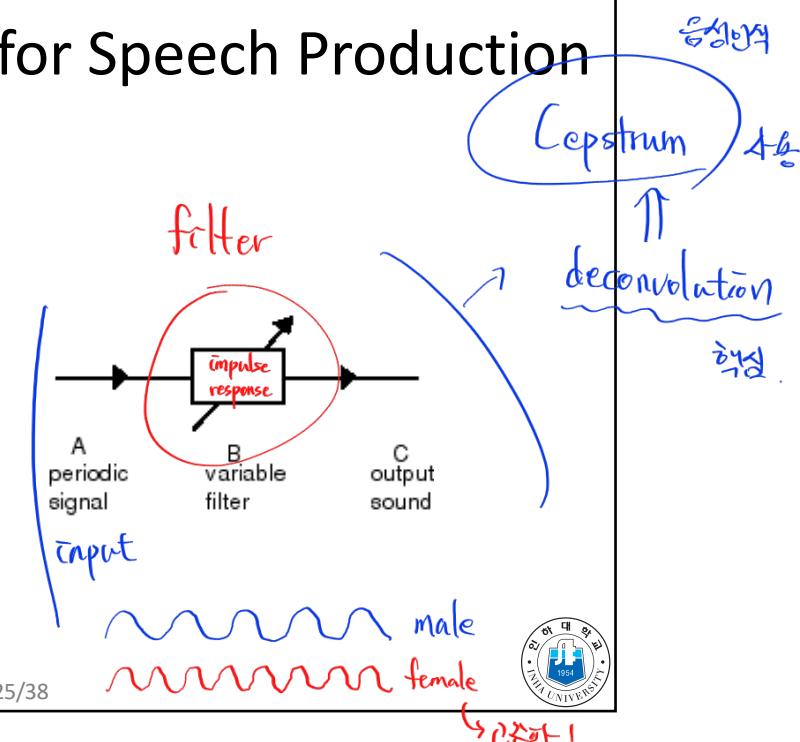
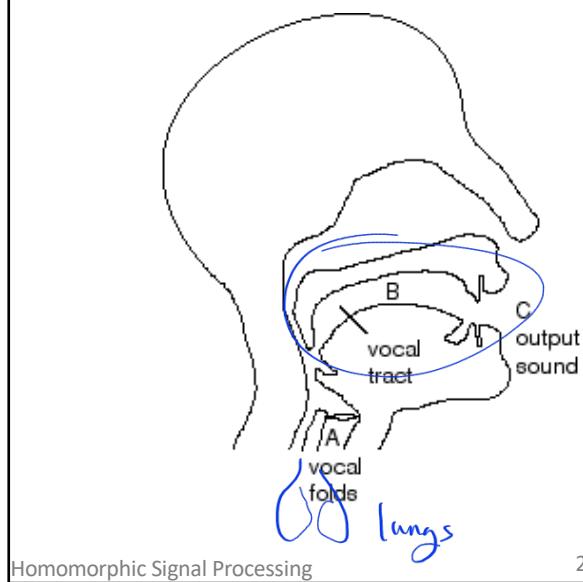


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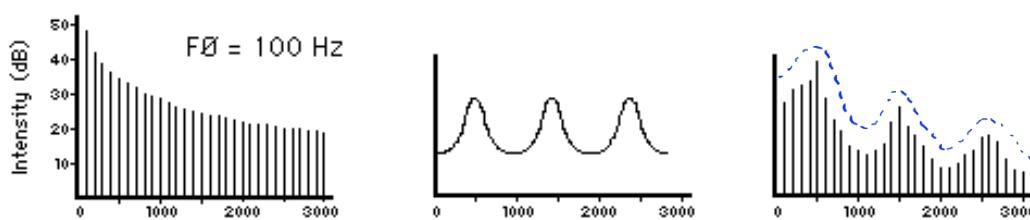


Source-Filter Model for Speech Production



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Source-Filter Model for Voiced Speech



Excitation

Vocal Tract

Output

$$E(e^{j\omega})$$

$$V(e^{j\omega})$$

$$X(e^{j\omega}) = E(e^{j\omega})V(e^{j\omega})$$

Logarithm of the Fourier Transform

- The output signal $x[n]$ is the convolution of
 - The excitation signal $e[n]$ and the vocal tract filter $v[n]$
- In the frequency domain

$$X(e^{j\omega}) = E(e^{j\omega})V(e^{j\omega})$$

- If we take the magnitude, then

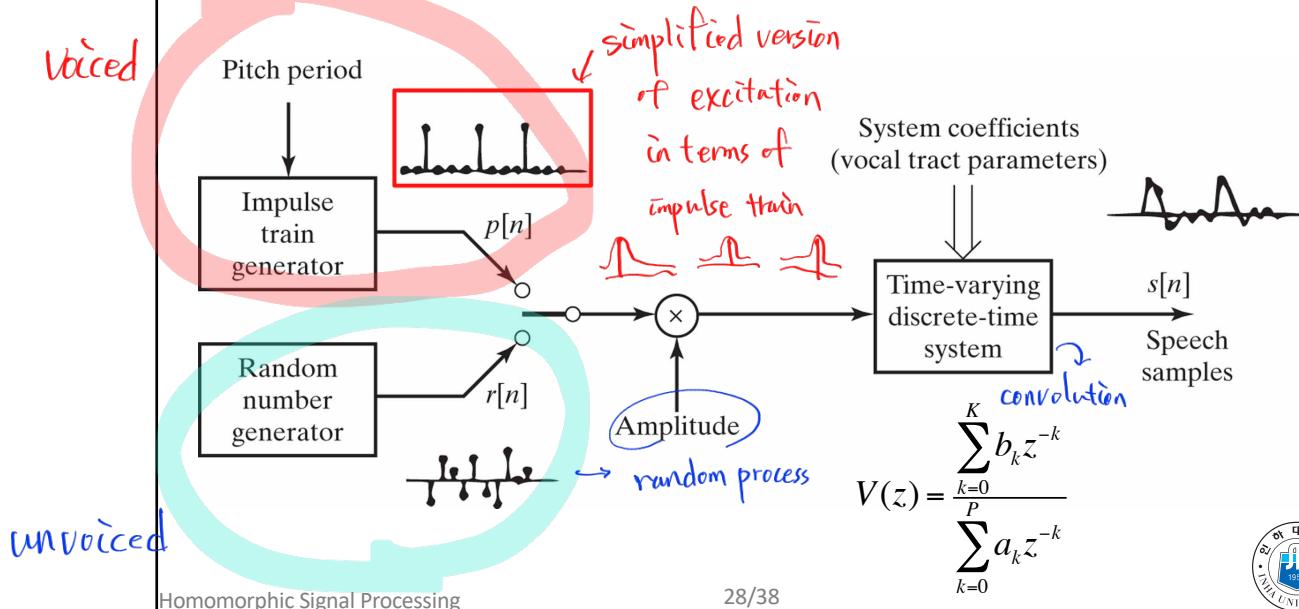
$$|X(e^{j\omega})| = |E(e^{j\omega})||V(e^{j\omega})|$$

- If we take the logarithm, then

$$\log|X(e^{j\omega})| = \log|E(e^{j\omega})| + \log|V(e^{j\omega})|$$



Speech Production Model



* noise? \Rightarrow unwanted signal.

즉, 흐릿한 것과 noise는 같은 것

Speech Deconvolution

- For voiced speech

$$s[n] = v[n] * p[n]$$

- After windowing

$$x[n] = \underbrace{w[n]}_{\text{windowing}} s[n] = \underbrace{v[n]}_{\text{windowing, to preserve this part.}} * \underbrace{p_w[n]}_{\text{excitation}}, \quad p_w[n] = w[n]p[n]$$

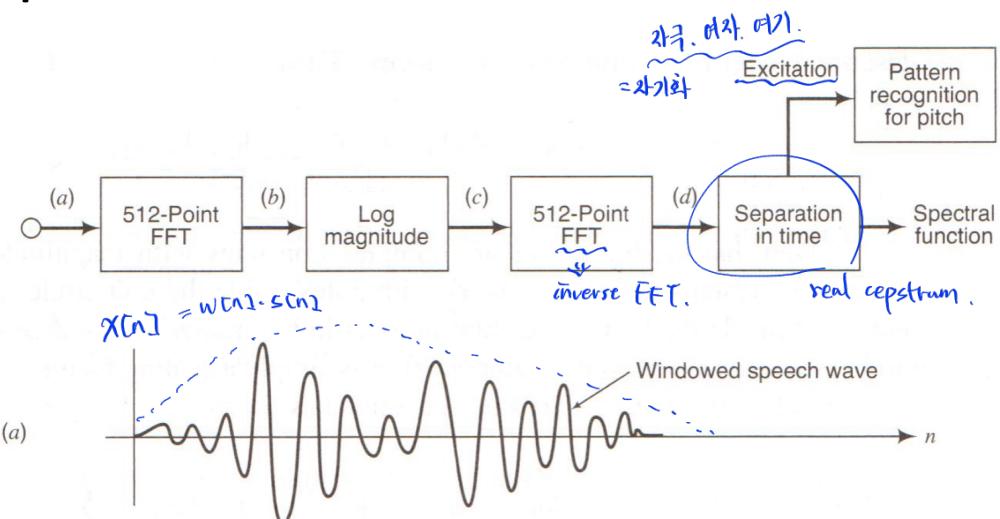
- The complex cepstrum

$$\hat{x}[n] = \hat{v}[n] + \hat{p}_w[n]$$

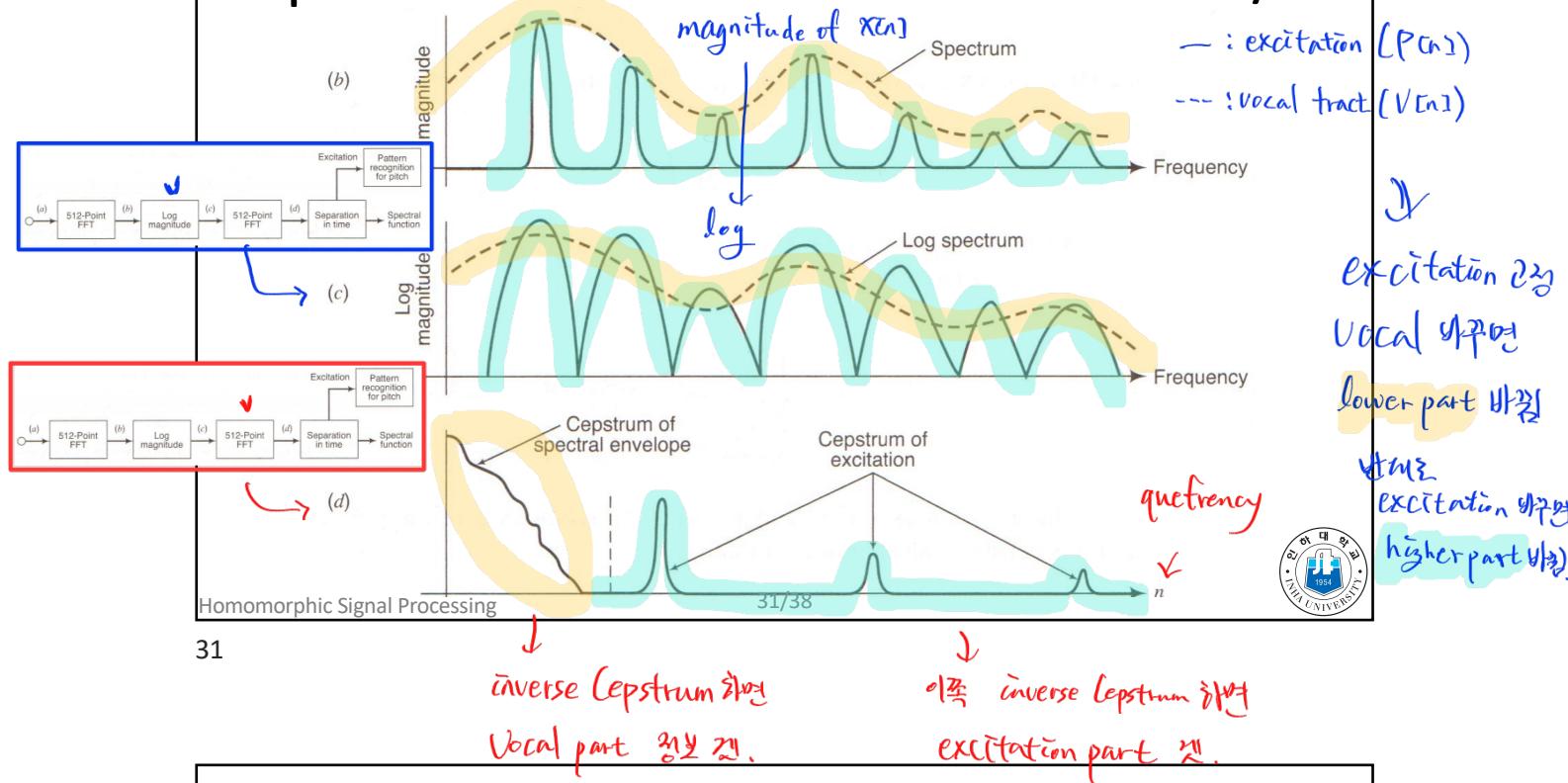
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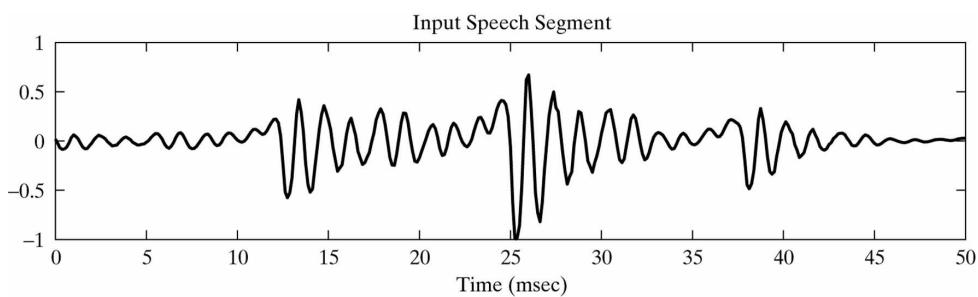
Speech Deconvolution Overview I/II



Speech Deconvolution Overview II/II



Speech Deconvolution Example I/V

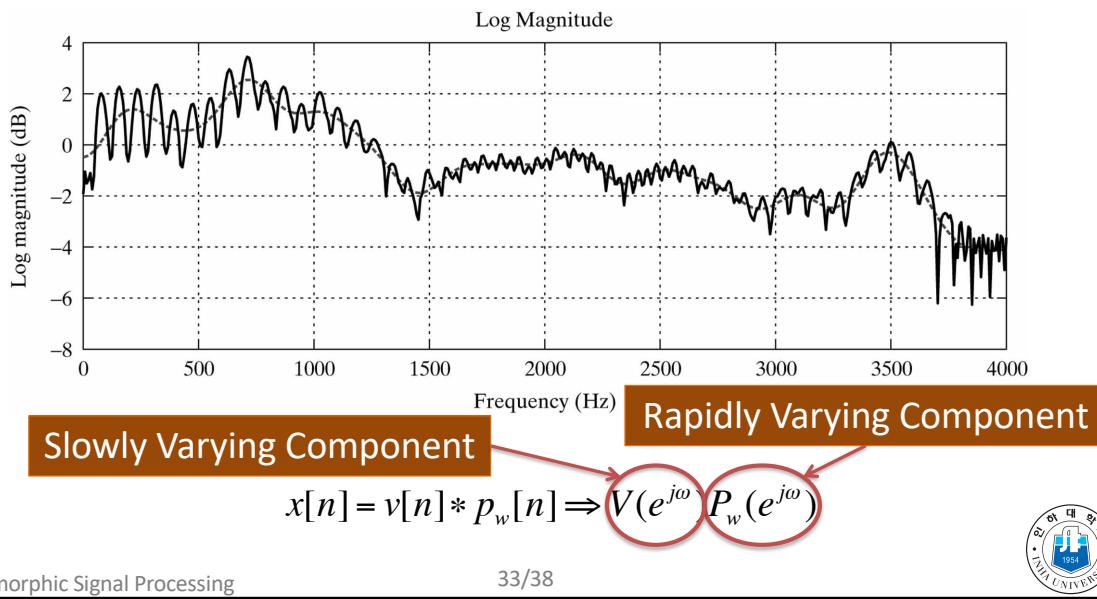


- Input speech signal after windowing

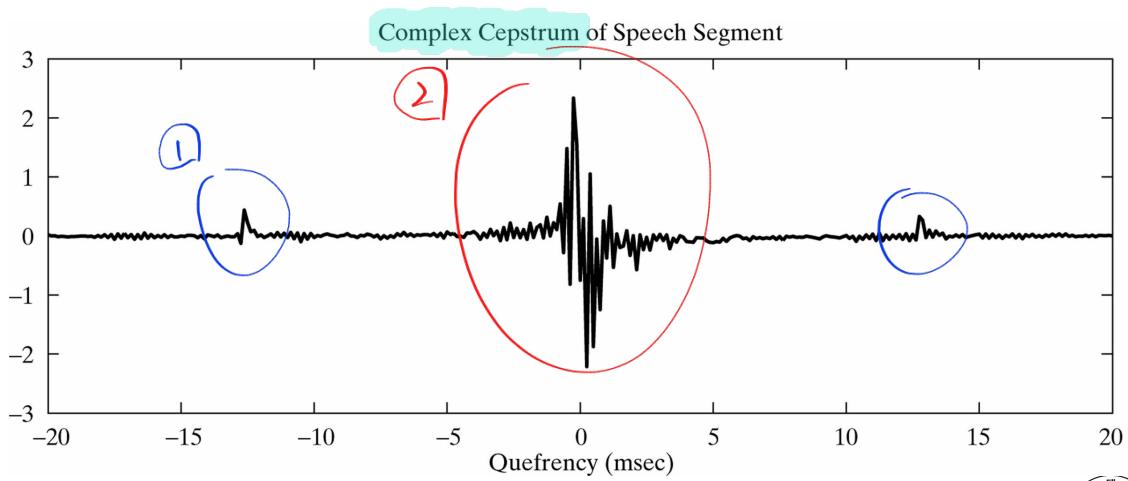
$$x[n] = w[n]s[n] = v[n] * p_w[n], \quad p_w[n] = w[n]p[n]$$



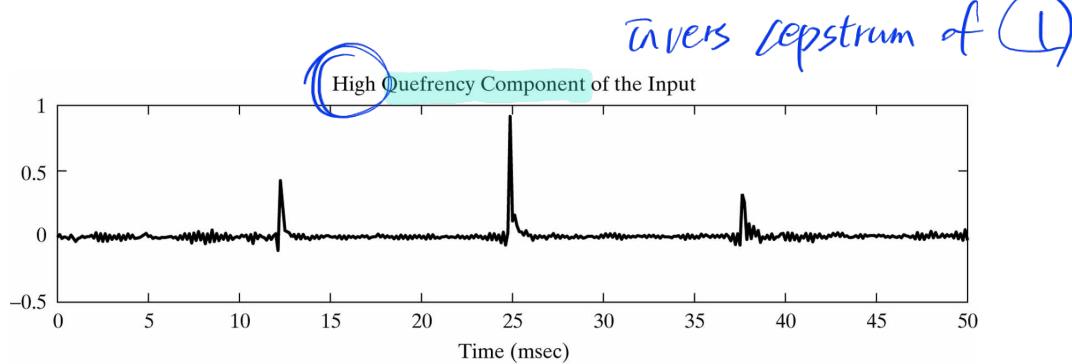
Speech Deconvolution Example II/V



Speech Deconvolution Example III/V



Speech Deconvolution Example IV/V

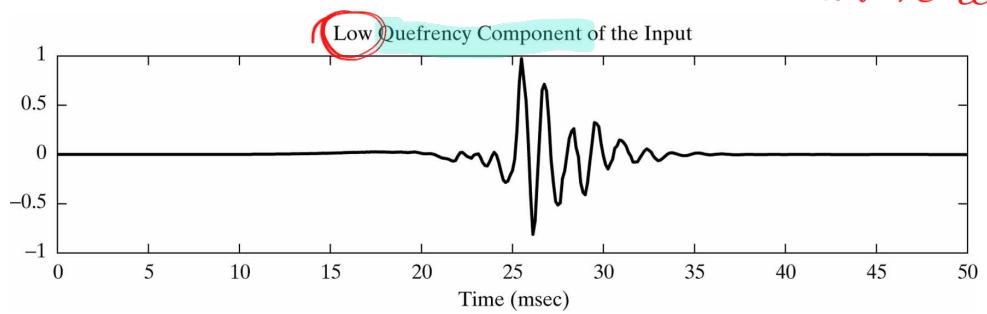


- High quefrency component

$$p_w[n] = w[n]p[n]$$



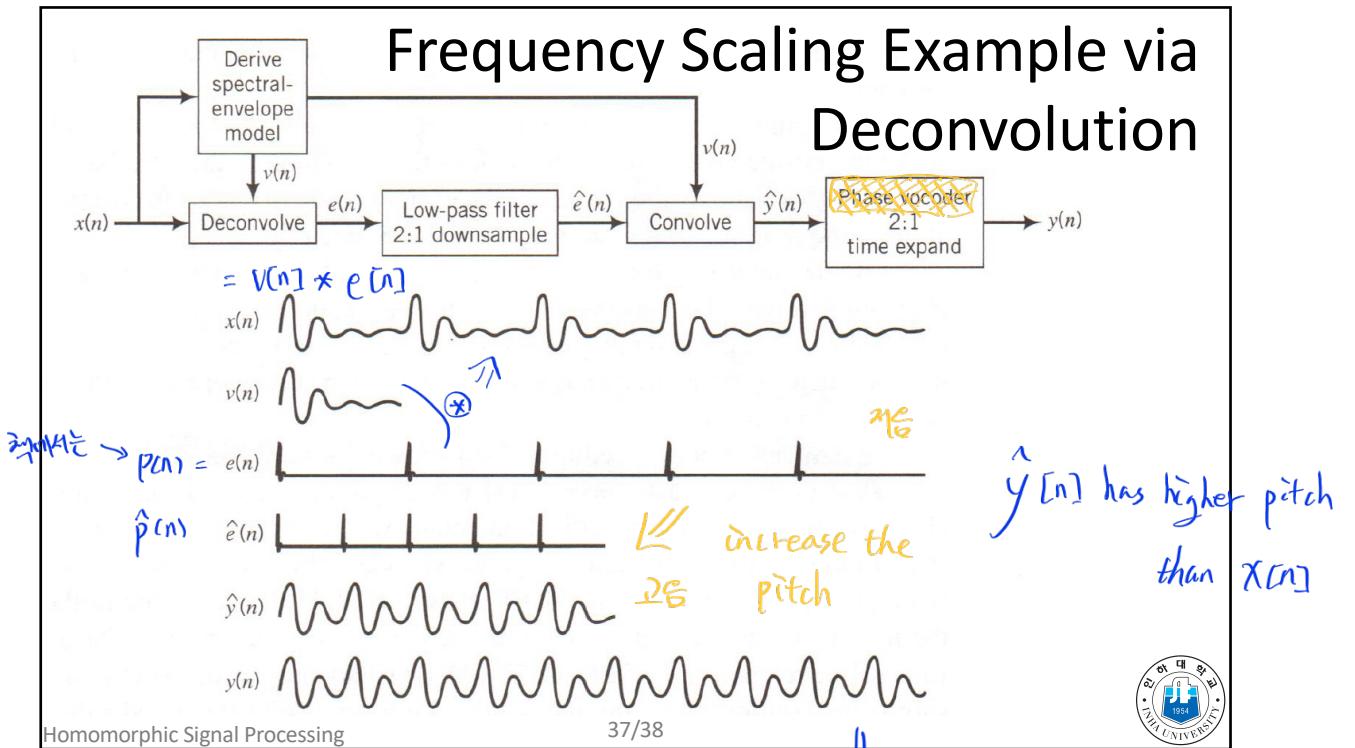
Speech Deconvolution Example V/V



- Low quefrency component

$$v[n]$$





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feature extraction → knockers

Questions?

Homomorphic Signal Processing

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