

# NVs for audio

Human perception:

16 Hz  $\rightarrow$  20 kHz

air pressure

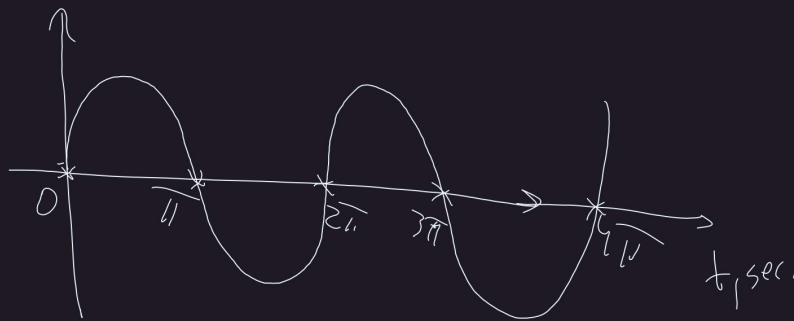


Wav-file:

bit rate 44.1 kHz

data width: 8-bit or 16-bit

number of channels (mono or stereo)



$2\pi$  sec.  $\rightarrow$  1 osc.

1 sec.  $\rightarrow \frac{1}{2\pi}$  osc.

$$B = \frac{1}{2\pi} \text{ Hz}$$

$$\frac{1}{2B} = \frac{2\pi}{2} = \pi$$

44 kHz  $\rightarrow$  16 kHz

Th (Nyquist-Shannon)

$f(t)$  has largest freq.  $B$  Hz  $\Rightarrow$

$f(t)$  can be restored from sampled signal,  
if it is sampled  $< \frac{1}{2B}$  samples per sec.

$$f(t) = A \sin(\omega t + \varphi)$$

$A$  - amplitude

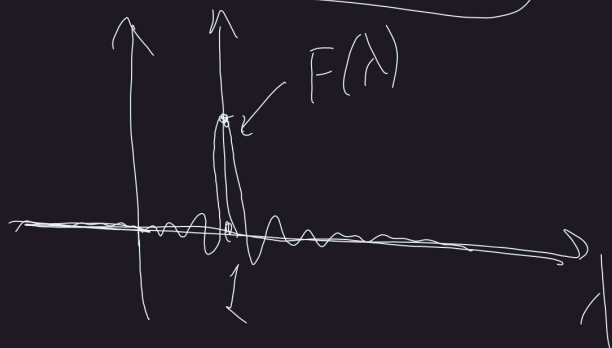
$\omega$  - freq.

$\varphi$  - phase



$$f(t) \longrightarrow A, \omega, \varphi$$

$$f(t) = \sin(t)$$



$$F(\lambda) := \int_{-\infty}^{+\infty} f(t) \sin(\lambda t) dt =$$

$$F(\lambda) = \int_{-\infty}^{+\infty} f(t) \sin(\lambda t) dt = \delta(\lambda - 1)$$

$$\approx \frac{1}{T} \sum_{n=-\infty}^{+\infty} f\left(\frac{n}{T}\right) \sin\left(\lambda \frac{n}{T}\right)$$

$$f(t) = \sin(t + \varphi)$$

$$F(\lambda) = \int f(t) \sin(\lambda t) dt$$



$$F(\lambda) = \int_{-\infty}^{\infty} f(t) \exp(-i\lambda t) dt$$

$$A = |F(\lambda)|^2$$

$$\varphi = \arg F(\lambda)$$

$$\begin{cases} F_1(\lambda) = \int f(t) \sin(\lambda t) dt \\ F_2(\lambda) = \int f(t) \cos(\lambda t) dt \end{cases}$$

$$A = F_1^2(\lambda) + F_2^2(\lambda)$$

doesn't depend on  $\varphi$

$$\varphi = \arctan\left(\frac{F_1(\lambda)}{F_2(\lambda)}\right)$$

Discrete Fourier Transform:

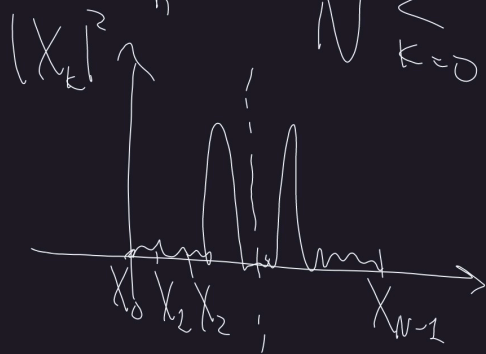
Complexity of DFT

$$x_0, x_1, \dots, x_{N-1} \xrightarrow{\text{DFT}} X_0, X_1, \dots, X_{N-1}$$

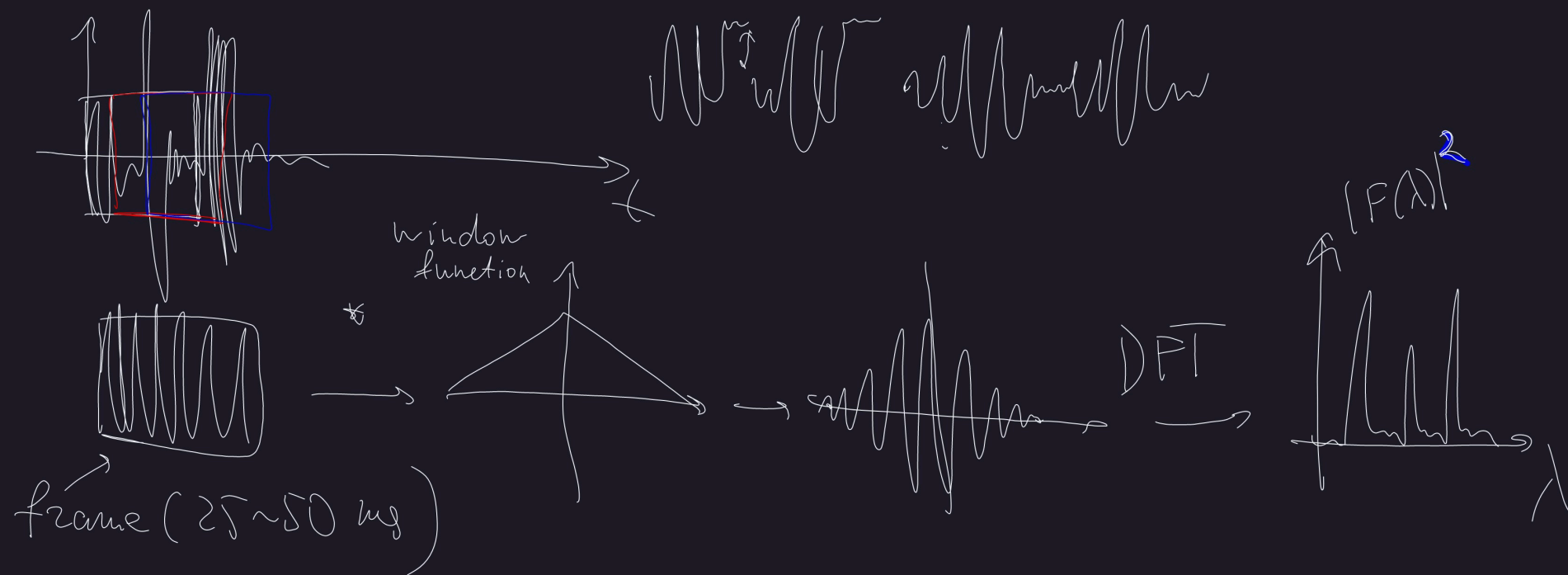
Direct comp.  $O(N^2)$   
FFT  $O(N \log N)$

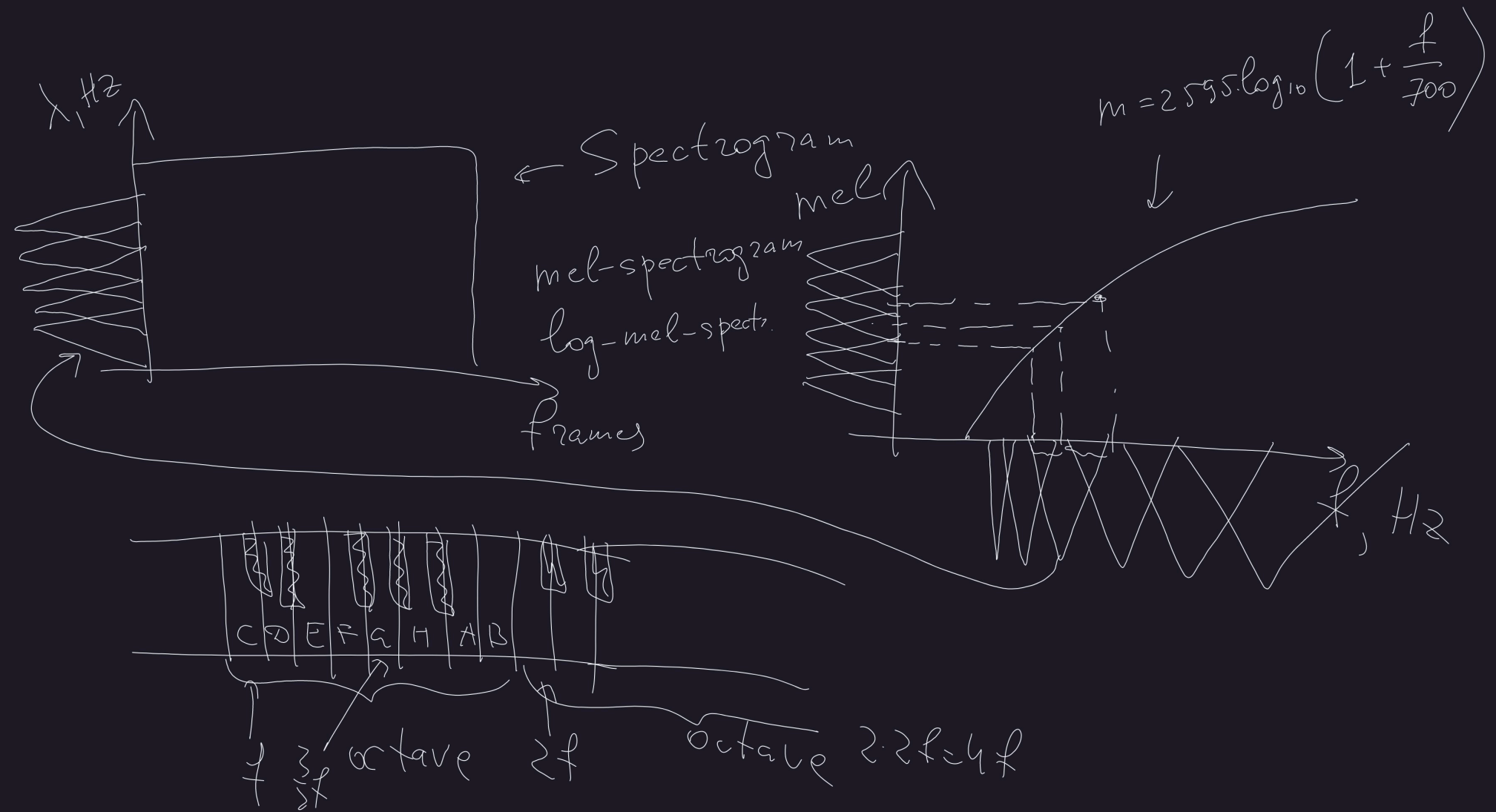
$$X_k = \sum_{n=0}^{N-1} x_n \exp\left(-i 2\pi \frac{k}{N} n\right) \quad \forall k=0, 1, \dots, N-1$$

$$x_n = \frac{1}{N} \sum_{k=0}^{N-1} X_k \exp\left(i 2\pi \frac{k}{N} n\right) \quad \forall n=0, 1, \dots, N-1$$



ASR: Audio  $\rightarrow$  Spectrogram  $\xrightarrow{NN}$  Text

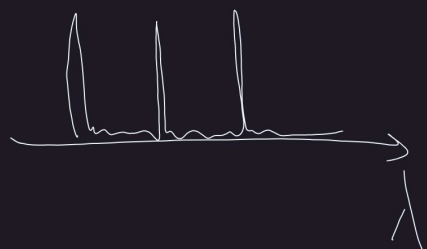




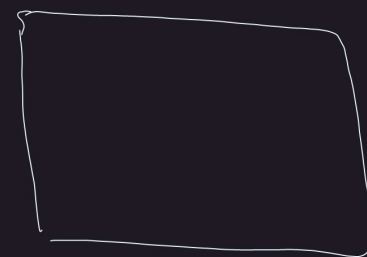
MFCC spectrogram

mel-frequency-cepstrum-coefs

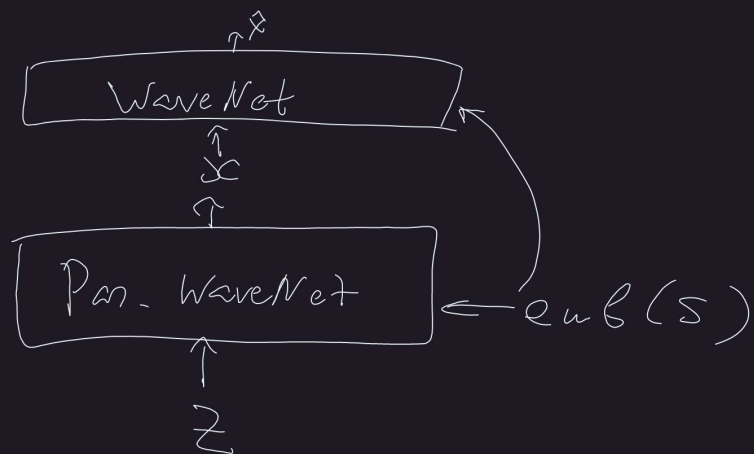
$$\text{MFCC} = \left| F \left( \log |F(f(t))|^2 \right) \right|^2$$



Audio →







$$x_t = z_t \cdot \sigma(z_{<t}) + \mu(z_{<t})$$

$$z \sim p(z) = \mathcal{N}(z|0, I)$$

$$\log p(x) = \log p(z) - \log \left| \det \frac{\partial x}{\partial z} \right| = \sum_t \log p(z_t) - \log \prod_{t=1}^T \sigma(z_{<t}) =$$

$$\left( \begin{array}{c|c} \Delta & 0 \\ \hline 1 & \end{array} \right) = \sum_t \log p(z_t) - \sum_t \log \sigma(z_{<t})$$

$$\| \text{KL} ( p_S(x|\theta) \parallel p_T(x) ) \rightarrow \min_{\theta}$$

$$\mathbb{E}_{p_S(x|\theta)} \log \frac{p_S(x|\theta)}{p_T(x)} = \mathbb{E}_{p_S(x|\theta)} \left( \sum_t \log p(z_t) - \sum_t \log \sigma(z_{<t}, \theta) \right) -$$

$$- \mathbb{E}_{p_S(x|\theta)} \sum_t \log p_T(x_t | x_{<t}) = \left\{ \begin{array}{l} z \sim p(z), x = f(z) \\ \mathbb{E}_{p(x)} g(x) = \mathbb{E}_{p(z)} g(f(z)) \end{array} \right\} =$$

$$= \mathbb{E}_{p(z)} \left( \sum_t \log p(z_t) - \sum_t \log \sigma(z_{<t}, \theta) \right) - \mathbb{E}_{p(z)} \sum_t \log p_T \left( \underbrace{z_t \cdot \sigma(z_{<t}) + \mu(z_{<t})}_{x_t} \parallel \dots \right)$$