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A Sub-band Filter Design Approach for Sound Field Reproduction

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A Sub-band Filter Design Approach for Sound Field Reproduction

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- Introduction
- Methodology
- Results
- Conclusions

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Statement of research problem

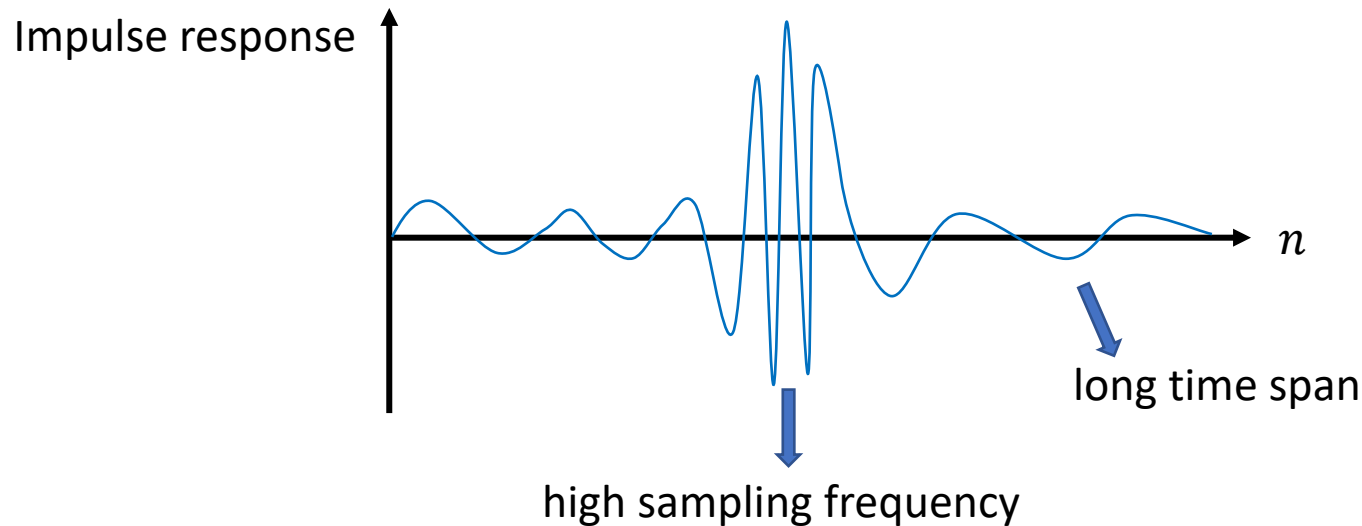
❑ Sound field reproduction uses loudspeakers to produce desired sound at locations.

❑ When designing filter for sound that spans a wide frequency range:

Low frequency band → longer time span

High frequency band → higher sampling frequency

→ Large number of filter coefficients



Statement of research problem

An approach is proposed to design filter in a sub-band form:

Design all sub-band filters directly in one optimization problem:

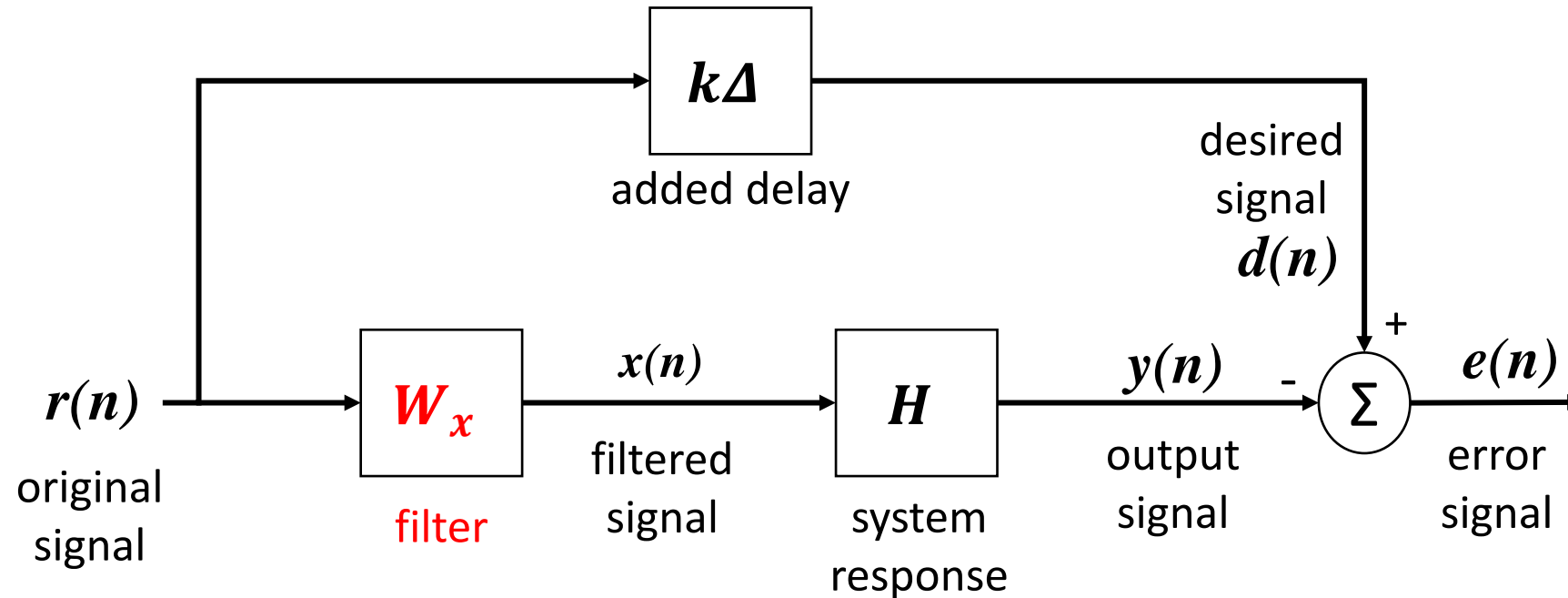
The **transition region** between two sub-band filters can be designed conveniently

The **computational load** can be reduced even if sub-band filters structure is not required

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Designing filter directly



❑ Example:

Use loudspeaker to produce desired sound at certain locations

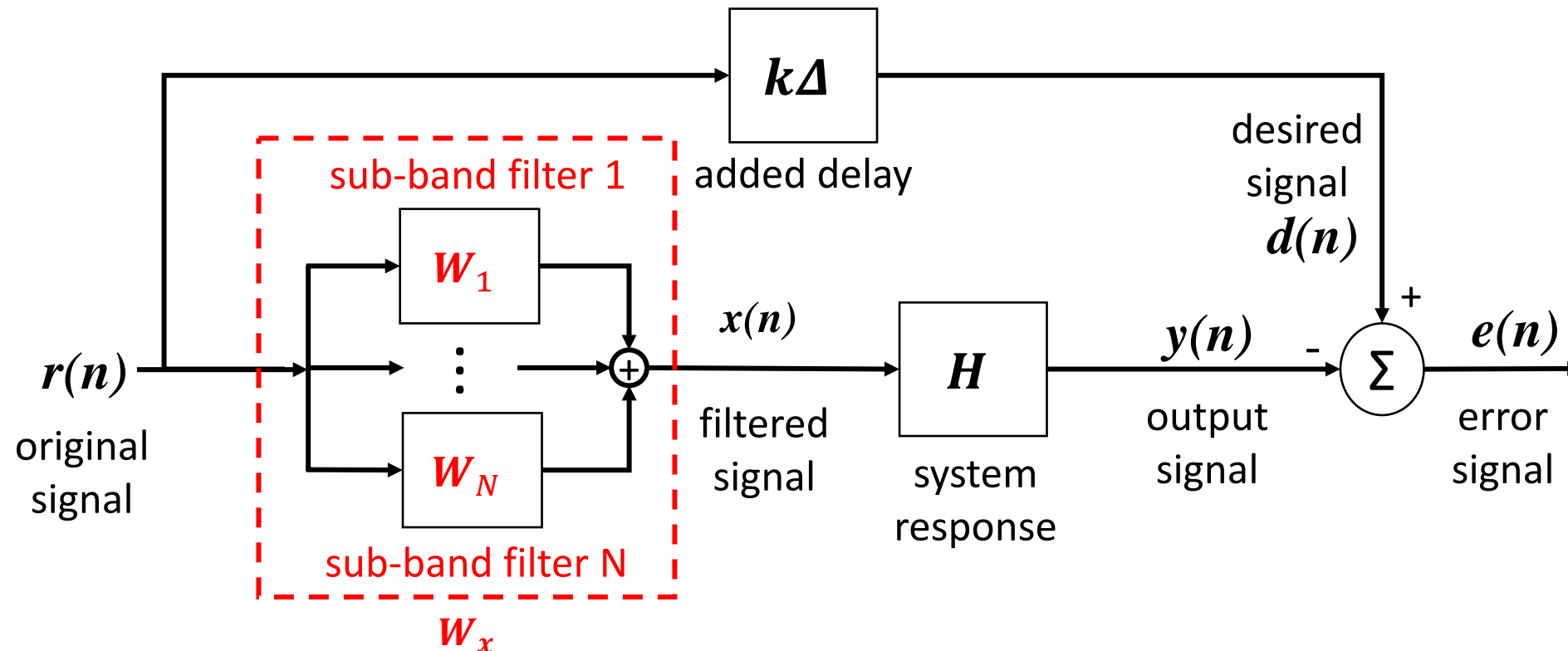
❑ Cost function:

Minimizing the power of error signal e

❑ Constraints:

Filter response $W_x(f)$

Designing filter when sub-band technique is used



Example:

Use loudspeaker to produce desired sound at certain locations

Cost function:

Minimizing the power of error signal e

Constraints:

Filter response $W_i(f)$

Expressing sub-band filters as one equivalent filter

❑ Conventional method (one single filter)

frequency response of designed filter at frequency f_k :

$$W_x(f_k) = F(f_k, f_s, N_t) \vec{w}_x, \quad F(f_k, f_s, N_t) = \left[1 \quad e^{-\frac{j2\pi f_k}{f_s}} \quad \dots \quad e^{-\frac{j2\pi f_k(N_t-1)}{f_s}} \right]$$

f_s is the sampling frequency,

N_t is the number of filter coefficients,

\vec{w}_x is the filter coefficients

❑ Sub-band structure

frequency response of designed filter at frequency f_k :

$$\sum_{i=1}^N W_i(f_k) = \sum_{i=1}^N \left[1 \quad e^{-\frac{j2\pi f_k}{f_{s_i}}} \quad \dots \quad e^{-\frac{j2\pi f_k(N_t-1)}{f_{s_i}}} \right] \vec{w}_i$$

Expressing sub-band filters as one equivalent filter

So designing **sub-band filters** can be treated as:

designing **one filter** \vec{w}_x with **modified Fourier matrix** $\tilde{F}(f_k)$

$$\tilde{W}_x(f_k) = \sum_{i=1}^N W_i(f_k) = \sum_{i=1}^N \left[1 \quad e^{-\frac{j2\pi f_k}{f_{s_i}}} \quad \dots \quad e^{-\frac{j2\pi f_k(N_t-1)}{f_{s_i}}} \right] \vec{w}_i = \tilde{F}(f_k) \vec{w}_x,$$

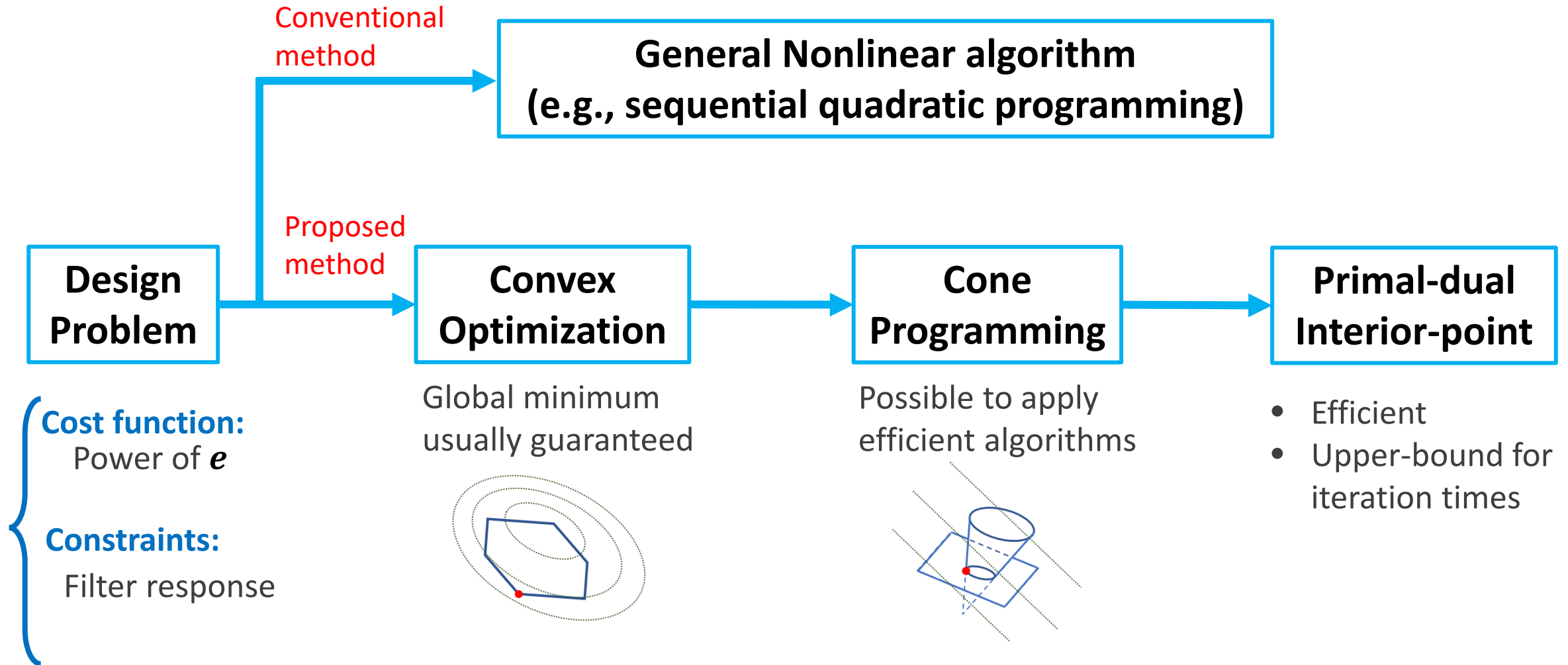
$$\tilde{F}(f_k) = \left[F(f_k, f_{s_1}, N_{t_1}) \quad \dots \quad F(f_k, f_{s_N}, N_{t_N}) \right],$$

$$\vec{w}_x = \begin{bmatrix} \vec{w}_1 \\ \vdots \\ \vec{w}_N \end{bmatrix}$$

So all the sub-band filters can be designed in **one optimization problem** if designed in the **frequency domain**.

The **transition region** can be designed more conveniently.

Overview of proposed design process



Problem formulation

Design Problem Expressed in Convex Problem

Cost function:

Total power of e:

$$\sum_{k=k_1}^{k_2} |E(f_k)|^2, \quad \longrightarrow \quad \vec{\tilde{w}}_x^T \left(\sum_{k=k_1}^{k_2} A_J(f_k) \right) \vec{\tilde{w}}_x + 2\text{Re} \left(\sum_{k=k_1}^{k_2} b_J^T(f_k) \right) \vec{\tilde{w}}_x + \sum_{k=k_1}^{k_2} c_J(f_k)$$

Convex ✓

- Quadratic
- $A_J(f_k)$ p.s.d

Constraints:

Filter response:

The magnitude of frequency response:

$$|W_i(f_k)| \leq C_i(f_k) \quad \longrightarrow \quad \|F(f_k, f_{s_1}, N_{t_1}) \vec{w}_i\|_2 - C_i(f_k) \leq 0 \quad \longrightarrow \quad \text{Vector norm} \quad \text{Convex} \checkmark$$

Cone Programming Reformulation

Convex Problem

Cost function:

$$\vec{w}_x^T \left(\sum_{k=k_1}^{k_2} A_J(f_k) \right) \vec{w}_x + 2\text{Re} \left(\sum_{k=k_1}^{k_2} b_J^T(f_k) \right) \vec{w}_x + \sum_{k=k_1}^{k_2} c_J(f_k)$$

Constraints:

$$\|F(f_k, f_{s_1}, N_{t_1}) \vec{w}_i\|_2 - C_i(f_k) \leq 0$$



Standard Cone Programming

Cost function: $c^T x$

Constraints: $x \in K_i, \quad i = 1, 2, 3 \dots$

$$Ax = b$$

c to be a constant vector

K_i to be a convex cone

A, b to be a constant matrix and vector

Cone Programming Reformulation

Convex Problem  Cone Programming




- Reformulate quadratic cost function

Cost function: $x^T A x + b^T x + c$



Cost function: $t_0 + b^T x$

Constraints: $\|\sqrt{A} x\|_2 \leq \sqrt{t_0 \tilde{t}_0}$
 $\tilde{t}_0 = 1$

-  Linear cost function
-  Rotated second-order cone
-  Linear constraint



- The vector norm constraint

Constraints: $\|x\|_2 - c \leq 0$



Constraints: $\|x\|_2 \leq t$

$t = c$

-  Second-order cone
-  Linear constraint

Cone Programming Reformulation

Convex Problem



Cone Programming

Cost function:

$$\vec{w}_x^T \left(\sum_{k=k_1}^{k_2} A_J(f_k) \right) \vec{w}_x + 2\text{Re} \left(\sum_{k=k_1}^{k_2} b_J^T(f_k) \right) \vec{w}_x + \sum_{k=k_1}^{k_2} c_J(f_k)$$

Constraints:

$$\|F(f_k, f_{s_1}, N_{t_1}) \vec{w}_i\|_2 - C_i(f_k) \leq 0$$

Cost function:

$$t_0 + 2\text{Re} \left(\sum_{k=k_1}^{k_2} b_J^T(f_k) \right) \vec{w}_x$$

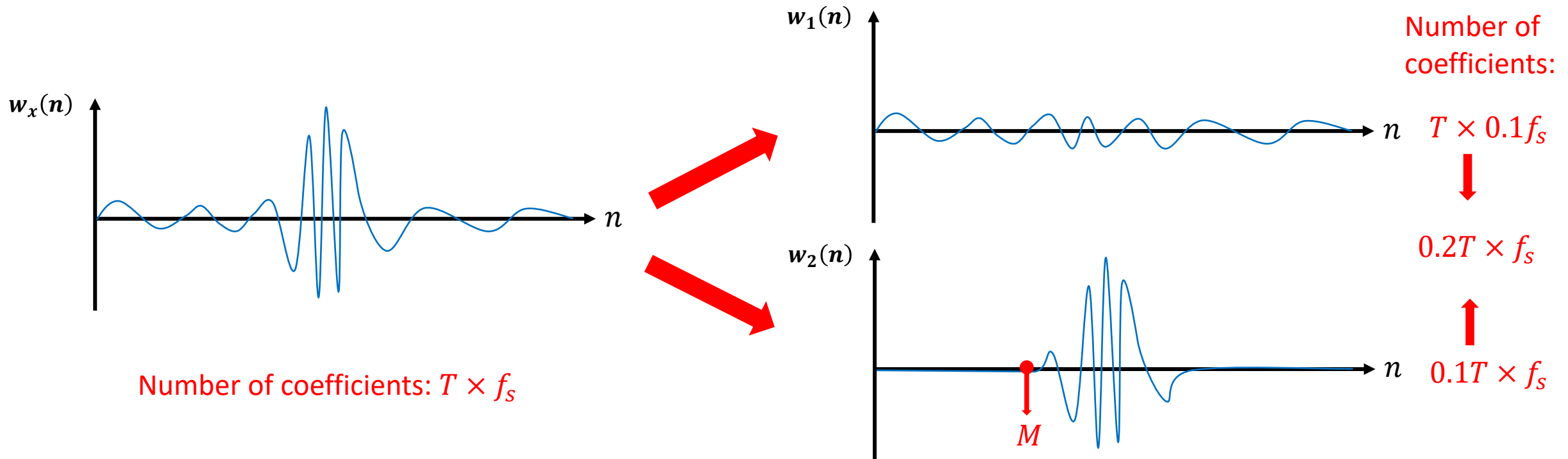
Constraints:

$$\|F(f_k, f_{s_1}, N_{t_1}) \vec{w}_i\|_2 \leq t_{3,k},$$

$$t_{3,k} = C(f_k)$$

A reduced order technique

Sometimes, the designed filter has high frequency response concentrated in small time span:



In this case, \vec{w}_i (with higher sampling frequency) can be chosen to start with $t = M\Delta$, where $M > 0$, then we have:

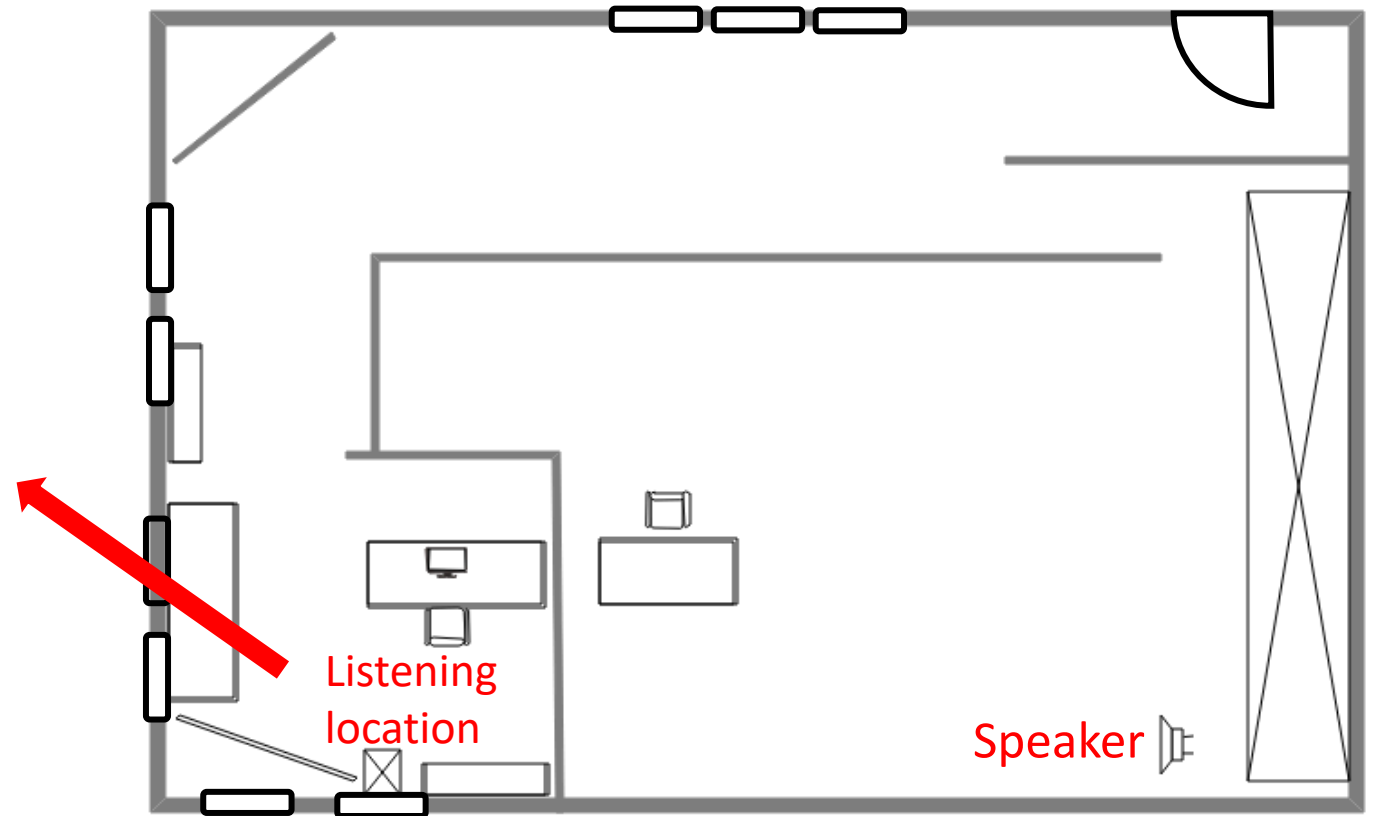
$$F_r(f_k, f_s, N_t) = \left[e^{-\frac{j2\pi f_k M}{f_s}} \quad e^{-\frac{j2\pi f_k (M+1)}{f_s}} \quad \dots \quad e^{-\frac{j2\pi f_k (N_t-1)}{f_s}} \right]$$

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Experimental setup

- An experimental setup for psychoacoustic listening test
- Speaker should produce desired sound at listening location



Experimental setup

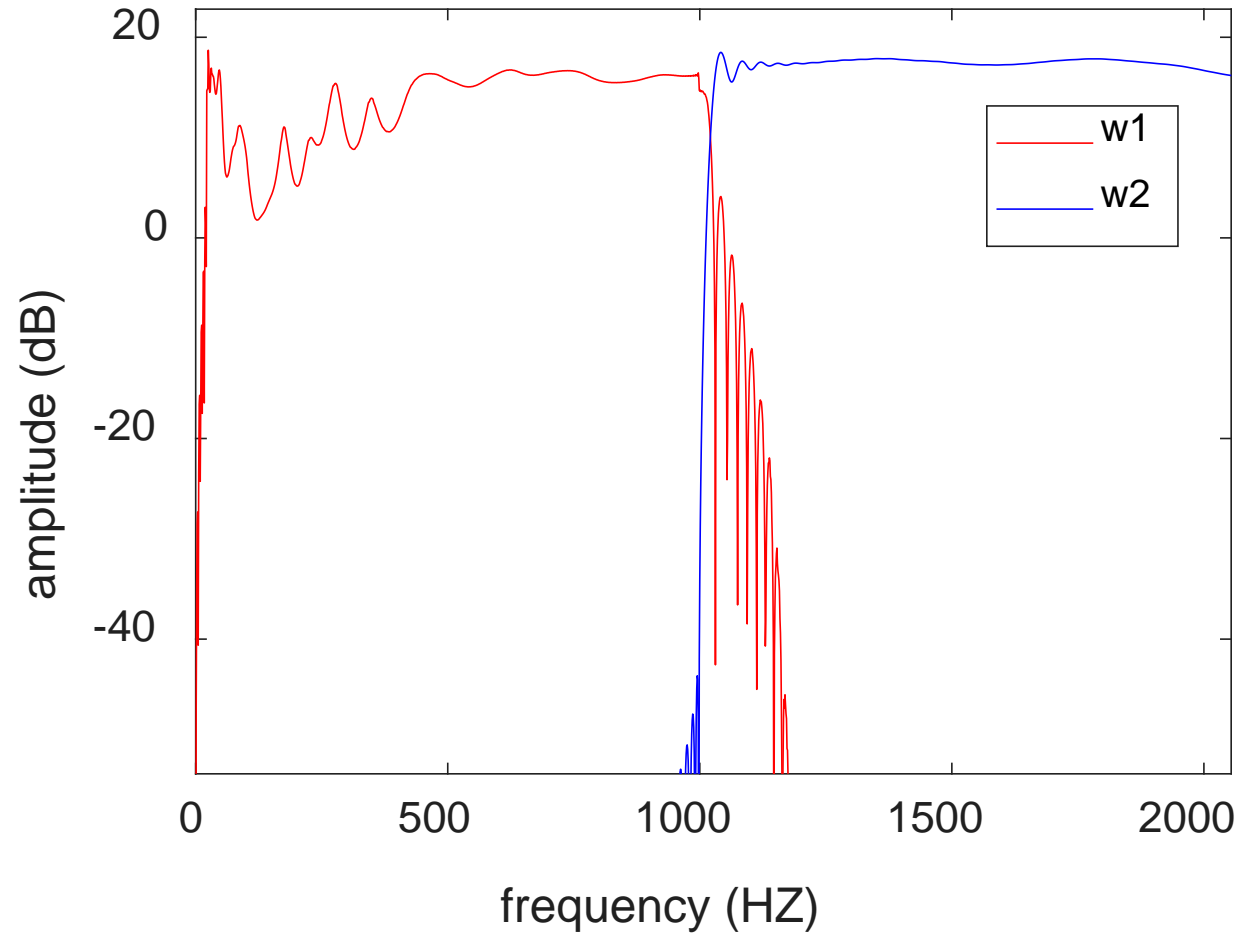
- Required sampling frequency: **48 kHz** ($\Delta = 20.83 \text{ us}$)
- Desired delay: **19200 Δ**
- Two sub-band filters:

	Sampling frequency	Filter coefficients	Starting time
Filter 1	2.4 kHz	1920	0
Filter 2	48 kHz	3000	17700 Δ

- SeDuMi is used to solve the reformulated cone programming problem

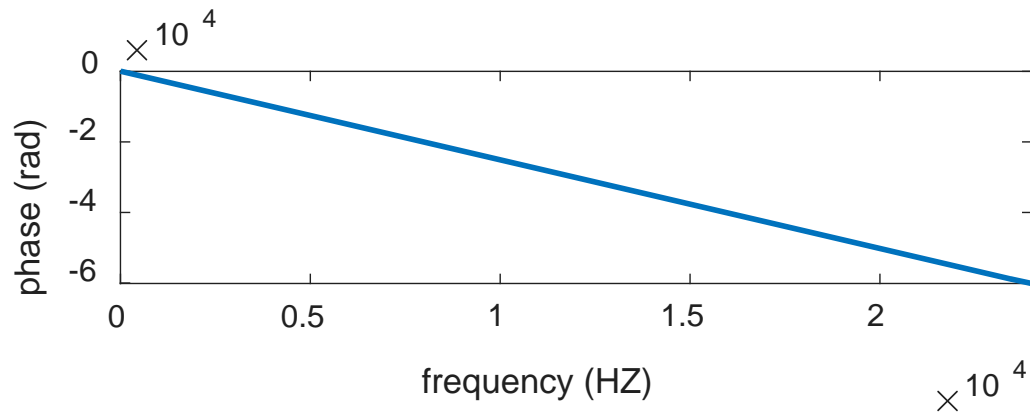
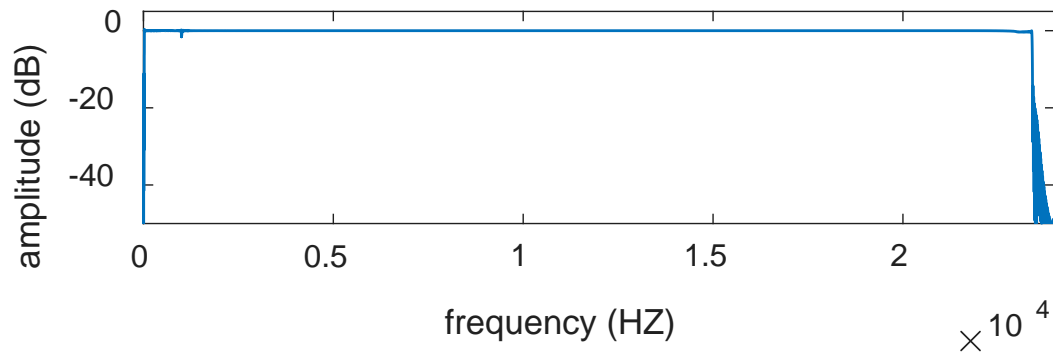
Result

The frequency response of both filter around 1200 Hz

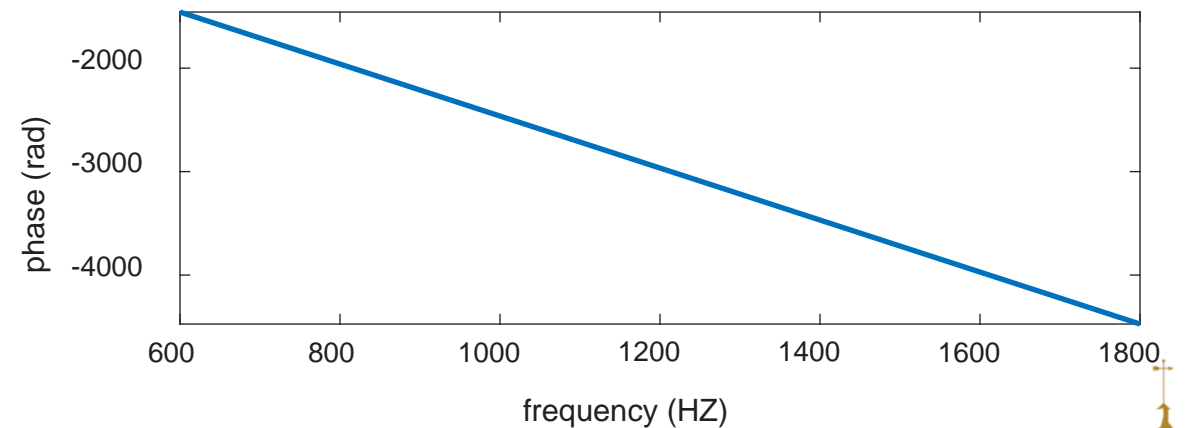
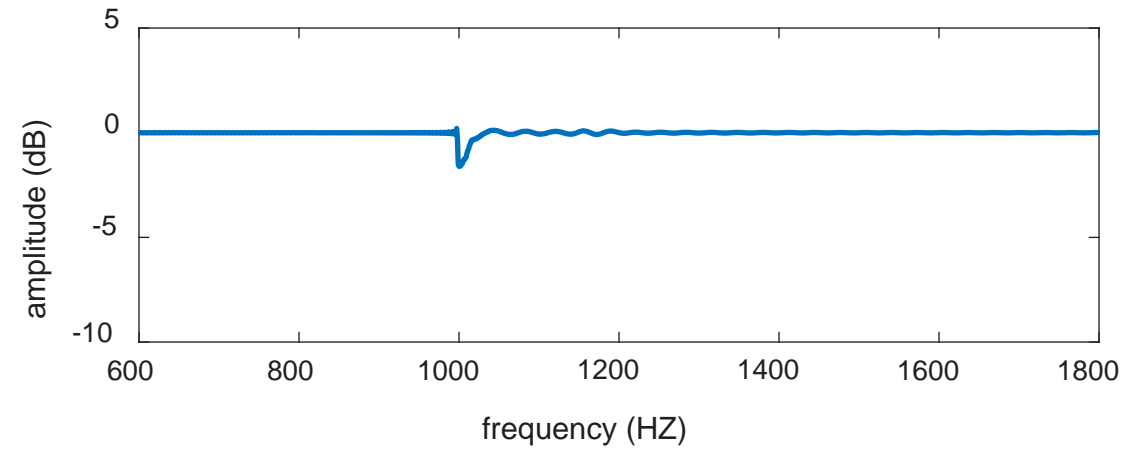


Result

The frequency response of $H(f)\tilde{W}_x(f)$



$H(f)\tilde{W}_x(f)$ around 1200 Hz

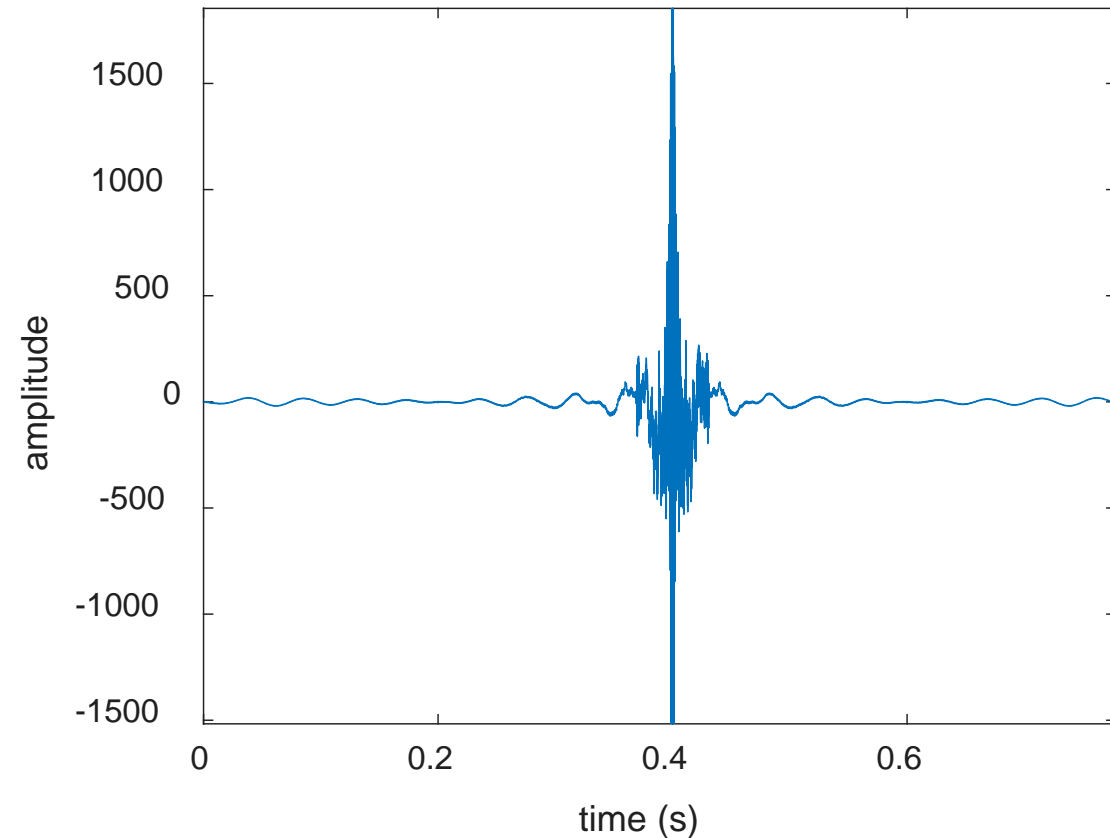


Result

Combining two sub-band filters together in time domain

The combination is done by:

- Upsampling the sub-band filter 1 with lower sampling frequency
- Adds the upsampled filter 1 with filter 2



The designed filter coefficients are $1920+3000 = 4920$, which is much smaller than $48000 \times \frac{1920}{2400} = 38400$

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Conclusions

- ❑ The proposed method can design sub-band filters for sound field reconstruction in one optimization problem, so designing transition region is more convenient.
- ❑ The optimization problem can be reformulated to a convex problem, then further reformulated to a cone programming problem. These guarantees the global optimal solution can be found in an efficient way.
- ❑ A reduced-order technique can be used to reduce the variables in filter design problem if different frequency bands of required filter have impulse response concentrated in different time intervals.

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