

# ROBUST AND LOW-COST CASCADED NON-LINEAR ACOUSTIC ECHO CANCELLATION

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# Acoustic Echo Approach

## Echo problem

- echo  $\implies$  loudspeaker signal picked up by the microphone
- echo and network delay  $\implies$  disturb communication

## Approach

Acoustic Echo  
Cancellation (AEC)

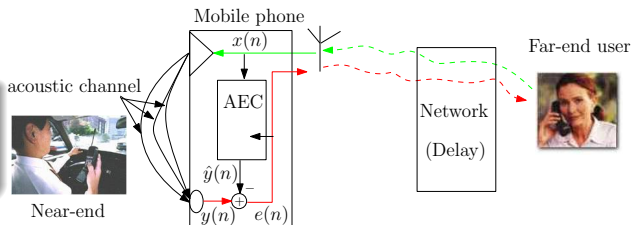


Figure: AEC approach

# Linear Acoustic Echo Cancellation

## characteristics

- linear environment:

$$y(n) = h(n) * x(n)$$

- non-linear environments:

~~$$y(n) = h(n) * x(n)$$~~

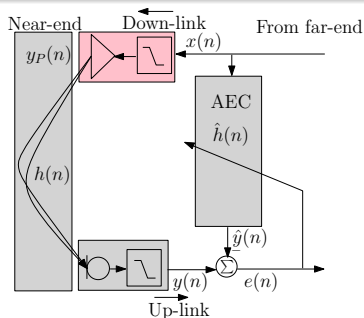


Figure: Non-linear loudspeaker

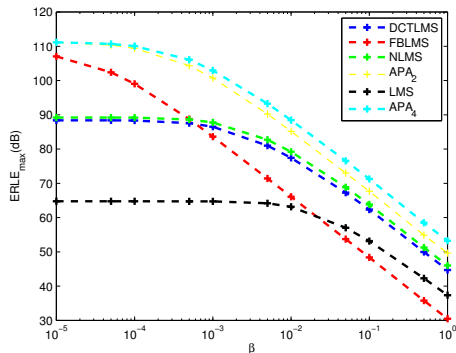


Figure: Loudspeaker non-linearity effect,  
 $y_P(n) = x(n) + \beta x^3(n)$

# Baseline Non-linear Acoustic Echo Cancellation

## characteristics

- non-linearity in acoustic path
- requires more computation capacity
- slow convergence
- depends on the model accuracy

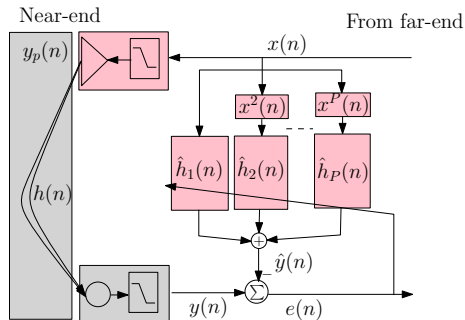


Figure: Non-linear AEC system

# Non-linearity Source and Model

## non-linear part

### loudspeaker

- non-linear filter
- small impulse responses
- slow variability

## linear part

### acoustic channel

- linear filter
- longer impulse response
- high variability

### microphone

- linear filter
- small impulse response
- low variability

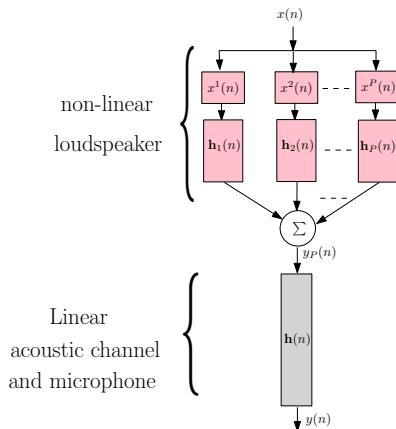


Figure: Proposed model

# Objectives

## proposed approach

- pre-processor  $\Rightarrow$  loudspeaker model [1, 2, 3]
- loudspeaker model  $\Rightarrow$  non-linear power filter [4, 5]
- power filter model from loudspeaker measurements [5]
- advantage: **robustness during acoustic channel changes**

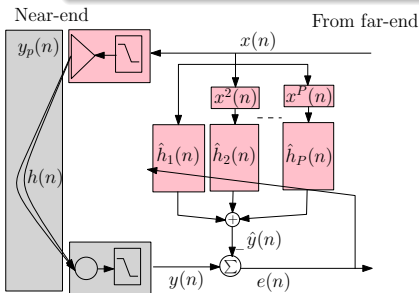


Figure: Parallel non-linear AEC approach

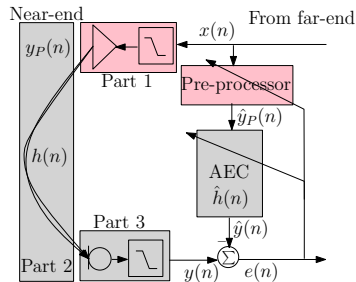


Figure: Cascaded non-linear AEC approach

# Non-linear AEC Approaches

## parallel or cascaded approach?

- $\tilde{h}_p(n) \implies \hat{h}_p(n) * \hat{h}(n)$
- parallel approach reach low minimum error
- parallel approach re-estimate all taps during **acoustic channel changes**

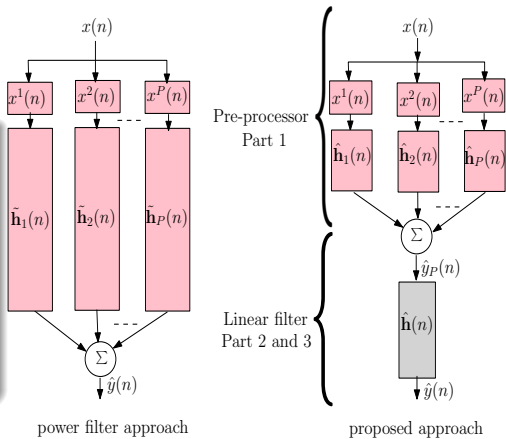


Figure: Parallel and cascaded approach



# Filter Estimation

## Signal Expression

- echo signal:

$$y_p(n) = \mathbf{h}_p(n)\mathbf{x}_p(n)$$

$$y(n) = \mathbf{h}(n) \sum_{p=1}^P \mathbf{y}_p(n)$$

- estimated echo signal:

$$\hat{y}_p(n) = \hat{\mathbf{h}}_p(n)\mathbf{x}_p(n)$$

$$\hat{y}(n) = \hat{\mathbf{h}}(n) \sum_{p=1}^P \hat{\mathbf{y}}_p(n)$$

- error signal:

$$e(n) = y(n) - \hat{y}(n)$$

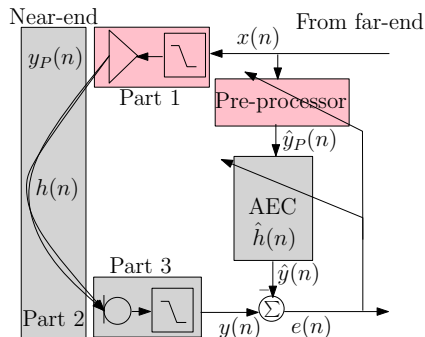


Figure: Cascaded non-linear AEC system

# Filters Estimation

## MMSE Estimator of Filters [6]

### AEC Linear Filter

$$\hat{\mathbf{h}} = \mathbf{r}_{y, y_P} \mathbf{R}_{y_P}^{-1} \quad (1)$$

$\mathbf{r}_{y, y_P}$ : cross-correlation

$\mathbf{R}_{y_P}$ : auto-correlation

### Comments

- depends on the loudspeaker signal
- $\mathbf{R}_{y_P} \Rightarrow$  determines convergence rate

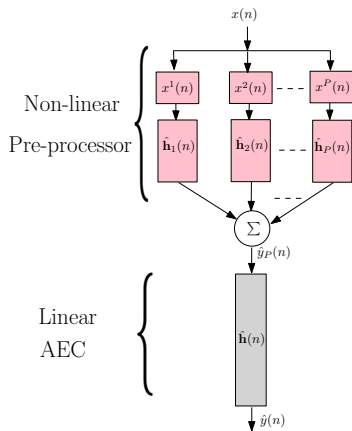


Figure: Proposed model

# Filters Estimation

## MMSE Estimator of Filters [6]

### Pre-processor Sub-filters

$$\hat{\mathbf{h}}_{p_k} = (\mathbf{r}_{y, \tilde{y}_{p_k}} - \mathbf{r}_{Y_{p \neq p_k}, \tilde{y}_{p_k}}) \mathbf{R}_{y_{p_k}}^{-1} \quad (1)$$

$$\tilde{y}_k(n) = x^{p_k}(n) * h(n)$$

$\mathbf{r}_{y, \tilde{y}_{p_k}}, \mathbf{r}_{Y_{p \neq p_k}, \tilde{y}_{p_k}}$  : cross-correlation

$\mathbf{R}_{y_{p_k}}$  : auto-correlation

### Comments

- $\mathbf{r}_{Y_{p \neq p_k}, \tilde{y}_{p_k}} \Rightarrow$  reduces performance
- correlation of input powers
- required of orthogonalization [4]

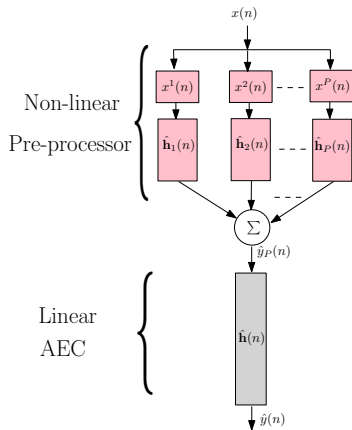


Figure: Proposed model

# Filters Estimation

Adaptive Estimation [1, 6]

## AEC Linear Filter

$$\hat{\mathbf{h}}(n+1) = \hat{\mathbf{h}}(n) + \mu \hat{\mathbf{Y}}_p(n) e(n)$$

### comments

- depends on the pre-processor output
- initialization:  $\exists p \setminus h_p(0) \neq 0$

## Pre-processor Sub-filters

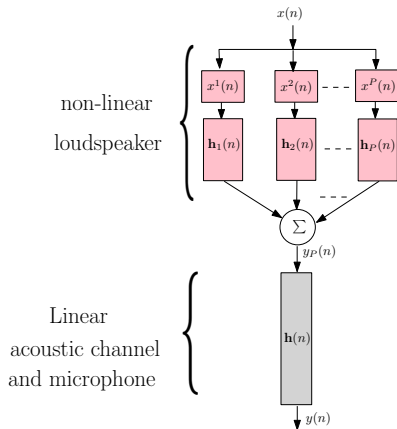
$$\hat{\mathbf{h}}_p(n+1) = \hat{\mathbf{h}}_p(n) + \mu_p \mathbf{X}_p(n) \hat{\mathbf{h}}^T(n) e(n)$$

### Comments

- depends on the estimate of the linear filter
- stability (lower step-size)

# Test Set-up

- pre-processor sub-filters  
 $P = 5$
- $\mathbf{h}_{p=1,2,3,4,5}(n)$ : 100 taps
- $\mathbf{h}(n)$ : 200 / 300 taps
- SNR ( echo signal to noise ratio ) 30 dB / 40 dB



## Echo Path Changes (EPC) (10s)

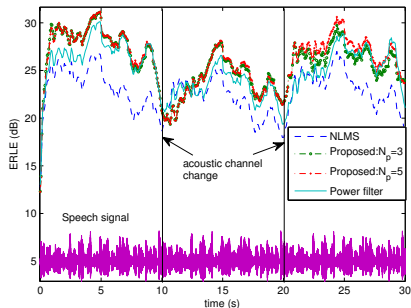
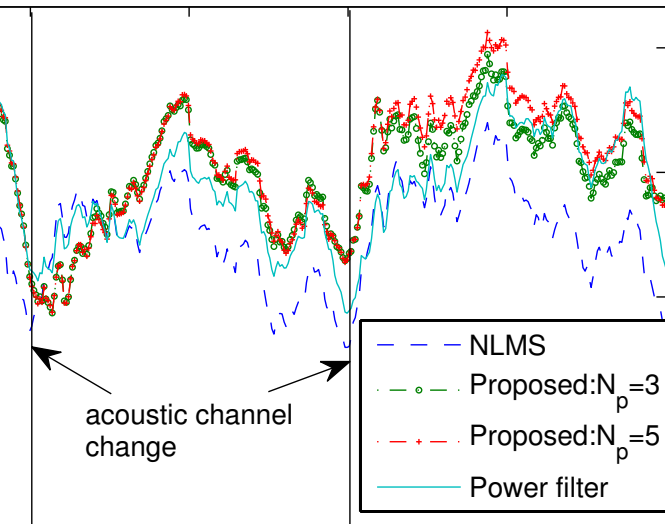


Figure: Comparison of ERLE, 200 taps, SNR=40 dB

### common parameters

- initial: better performance
- 1<sup>st</sup> EPC: pre-processor less accurate
- 2<sup>nd</sup> EPC: pre-processor more accurate, better ERLE for  $N_5$  than  $N_3$
- proposed model re-converge faster and become more robust to path changes

# Echo Path Changes (EPC) (10s)

1<sup>st</sup>

ac

2<sup>nd</sup>

ac

N

## Echo Path Delay Changes (EPC:10s, delay:2.5ms)

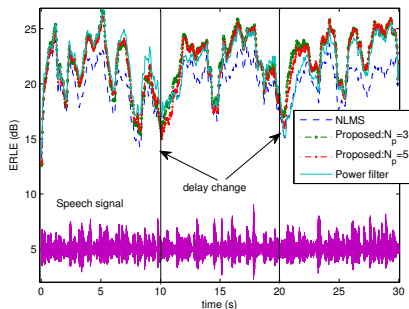


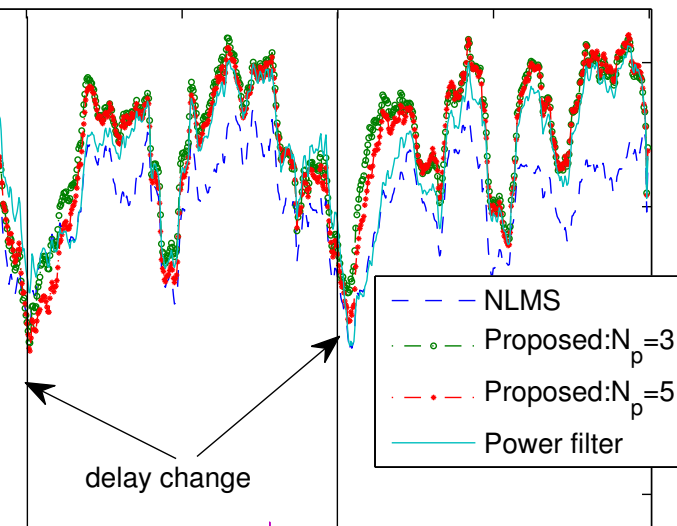
Figure: Comparison of ERLE, delay change, 300 taps, SNR=30 dB

### Best tests case

- initial: Similar performance
- 1<sup>st</sup> EPC: pre-processor not accurate
- 2<sup>nd</sup> EPC: pre-processor more accurate
- EPC improve the pre-processor accuracy



# Echo Path Delay Changes (EPC:10s, delay:2.5ms)



- 1<sup>st</sup>
- ac
- 2<sup>nd</sup>
- ac
- E

# Conclusions

## Conclusions

- cascaded approach to N-AEC
- robustness against path changes
- robustness in delay changes

## Perspectives

- reduce complexity
- pre-processor control with adaptive step-size

END

*THANKS FOR YOUR ATTENTION*

## References



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