

Reverberation Enhancement Systems with Time-Varying Mixing Matrices

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1. Introduction

2. Recap: Reverberation Enhancement Systems

- 3. Time-varying Processing
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- 5. Conclusion



What is a reverberation enhancement system?

Reverberation is added by controlled feedback between microphones and loudspeakers.





Why use a reverberation enhancement system?

Music venues may need

- improvement of poor acoustical design,
- flexible acoustics for different genres of music.

Reverberation enhancement systems (RESs) can

- increase the level of reverberation,
- extend the reverberation time,
- increase the sense of envelopment.



Electro-Acoustic Feedback: Isn't this dangerous?

The requirements for RESs are

- stability,
- absence of coloration.

The stability of RESs can be improved by

- de-coupling of loudspeakers and microphones,
- equalization,
- active feedback cancellation,
- time-variation.



Types of Time-Variation

Existing techniques are

- frequency shifting (FS) [Schroeder (1962)],
- delay line modulation (DL) [Griesinger (1991)],
- allpass modulation (AP) [Lokki and Hiipakka (2001)].

But

- frequency of musical signals can be altered considerably (FS),
- inaccurate frequency response and reverberation time (DL),
- time-variation itself can be unstable (AP),
- number of time-varying coefficients is small (AP, FS, DL).

The proposed technique is **feedback matrix modulation**, which is guaranteed to be stable and has a quadratic number of time-varying coefficients.



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Recap: Reverberation Enhancement Systems General Structure





Recap: Reverberation Enhancement Systems Stability of RES

The stability of an RES depends on $G_{ML}H_{LM}$.

Nyquist stability for single channel feedback:

G_{ML}(ω)**H**_{LM}(ω) does not encircle the point 1 + i0

Nyquist stability for multi channel feedback [MacFarlane and Postlethwaite (1977)]:

Given the eigenvalue decomposition per frequency:

$$\mathbf{G}_{ML}(\omega)\mathbf{H}_{LM}(\omega) = \mathbf{V}\,\boldsymbol{\Gamma}(\omega)\mathbf{V}^{-1},$$

• Characteristic functions $\gamma_i(\omega)$ do not encircle the point 1 + i0.

Recap: Reverberation Enhancement Systems Gain-Before-Instability





The gain-before-instability (GBI) is the gain μ where $\mu G_{ML} H_{LM}$ has a 50% risk of instability.

 ${\cal K}$ is the number of independent frequency bins.





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Time-varying Processing Why does time-variation help? 5 5 Magnitude [dB] Magnitude [dB] 0 0 -5 -10 -10 -15 -15 -20 -20 1.02 1.04 1.06 Frequency [kHz] 1.08 1.1 1.02 1.06 1.08 1 1.04 1.1 Frequency [kHz] time point 1 time point 2 10 Magnitude [dB] -2-10-12 5 -20 1.04 1.06 Frequency [kHz] 1.02 1.08 1.1 1

mean frequency response



Time-varying Processing Two Types of Processors





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Time-varying Processing Time-Varying Mixing Matrix



The eigenvalue decomposition of the feedback matrix is given by

$$\mathbf{A}(n) = \mathbf{U}^H \, \mathbf{\Lambda}(n) \, \mathbf{U},$$

where $\Lambda(n) = \text{diag}(\lambda_1(n), \dots, \lambda_N(n))$ with the eigenvalues $\lambda_i(n)$. Matrix $\mathbf{A}(n)$ is real, but \mathbf{U} and $\Lambda(n)$ are typically complex.

For energy conservation, all $|\lambda_i(n)| = 1$, i.e. all eigenvalues lie on the unit circle.

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Time-varying Processing

Time-varying Matrix Eigenvalues

The angular evolution of the eigenvalues are triangular waves with

- modulation frequency,
- modulation amplitude,
- modulation spread.

Starting values can be chosen arbitrarily, e.g. all as zeros.



 $\Rightarrow N$ time-varying eigenvalues, but N^2 time-varying matrix entries.



Time-varying Processing

Time-Varying Feedback Delay Network (FDN)



- Feedback loop creates artificial reverberation with a specified reverberation time.
- Optimized structure to maximize effect of time-variation.



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Evaluation

Choice of Simulation Room

- Small office space with size 5.8 x 3.9 x 3.1 m
- Mean reverberation time of 0.5 s
- 4 loudspeakers and 4 microphones
- Three types of room acoustics



Evaluation Choice of Processor

Matrix Modulation has

- modulation amplitude 0-2,
- mean frequency up to 3 Hz,
- frequency spread of 10%.

Feedback Delay Network has

- 8 delay lines,
- reverberation time of 1.5 seconds.



Evaluation GBI with Time-Varying Mixing





Evaluation GBI with Time-Varying FDN





Evaluation Comparison between Mixing and FDN





Evaluation GBI Improvement via Time-Variation

For the measured room impulse responses:

GBI [dB]	FDN	Mixing	
No Time-Variation	-13.1	-9.7	
Max Time-Variation	-5.8	-6.6	
Improvement	7.3	3.1	

 \implies The enhancement of 7.3 dB is an improvement of 30% compared to the 5.6 dB found for time-varying allpass FDNs.



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Conclusion

Take home messages

Time-varying mixing matrices in RESs have been proposed:

- Two types of processor: mixing and FDN.
- RES optimized FDN structure has been proposed.
- Quadratic number of time-varying coefficients.

Simulation results show

- GBI increases up to 7.4 dB for FDN based RES.
- GBI improvements of 30% compared to allpass FDNs.



Conclusion Open Questions

- What is the perceptually preferable time-variation technique?
- What is the perceptually preferable amount of modulation?



Conclusion

Thank you very much for your attention.



Conclusion Two Types of Processors

Mixing Matrix

- Relies on microphone loudspeaker coupling
- Computationally cheap
- Reverberation time is dependent on feedback gain
- RES extends the physical reverberation

Feedback Delay Network

- Independent from mic loudspeaker coupling
- Computationally expensive
- Reverberation time is specified by FDN
- RES superposes artificial reverberation



Conclusion Risk of Instability without Time-Variation





Conclusion

Comparison to Time-Varying Allpass FDN

Time-varying Allpass FDNs [Lokki and Hiipakka (2001)]

- can be unstable under strong variation
- introduce a frequency dependent delay which decreases the accuracy of the reverberation time specification
- with comb-based allpass filters have ringing tails, which depend on the feedforward-feedback gain
- there are only two time-varying coefficients per delay line
- provide experimental GBI improvement up to 5.6 dB.



Conclusion Extrema of the mean GBI

Mixing	Min[dB]	Max[dB]	Enhancement [dB]
Anechoic	-3.8	-1.2	2.6
Measured	-9.7	-6.6	3.1
Synthetic	-4.9	-2.0	2.9

FDN	Min[dB]	Max[dB]	Enhancement [dB]
Anechoic	-7.6	-1.1	6.5
Measured	-13.1	-5.8	7.3
Synthetic	-8.5	-1.1	7.4

 \implies Improvement of 30% compared to time-varying allpass FDN.



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