

A?
Aalto University
School of Electrical
Engineering

ELEC-E5640 Noise Control
27 Nov. 2023



Sound check 


Active Noise Control

Prof. Vesa Välimäki
Acoustics Lab, Dept. Information and Communications Engineering,
Aalto University

1

Active Noise Control – Contents

- Fundamentals of active noise control
- Adaptive filtering
- Interference cancellation
- Applications
 - Ventilation duct, headphones, transportation



A? Aalto University
School of Electrical
Engineering

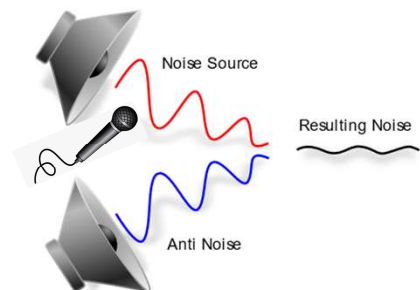
© 2006-2023 Vesa Välimäki

2

2

Active Noise Control Basics

- Electronics is faster than sound!
 - Detect noise with a microphone
 - Produce anti-noise
 - Play it with a speaker
 - Noise and anti-noise will cancel out
- Advantage: good attenuation at low frequencies
- Acoustic transfer function modeling
 - How does noise propagate from mic to the area to be silenced?
- Size of silent area depends on geometry
 - In 1-D case: perfect, in 3-D case: small



ANC = Active Noise Control

https://en.wikipedia.org/wiki/Active_noise_control

3

A Short History of ANC

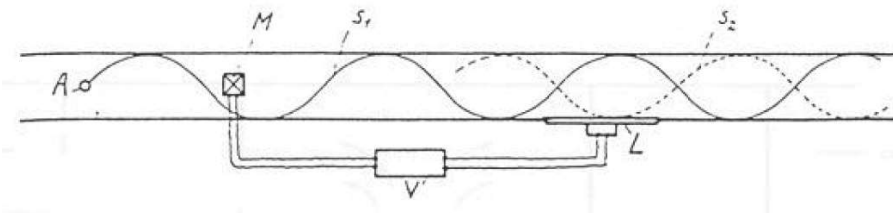
- Paul Lueg's invention in Germany in the 1930s
- Practical implementation become possible only decades later
 - Manual systems in the 1950s
 - Automatic control of analog ANC systems in the 1960s
 - Digital ANC systems in the 1980s
- First textbook in 1992
 - P. A. Nelson and S. J. Elliott (ISVR, Univ. of Southampton, UK),
"Active Control of Sound"
- Industrial applications appeared in the 1990s
 - Active hearing protectors, attenuators for ventilation ducts, household appliances, cars, aircrafts



4

Principle of ANC in a Duct

- Paul Lueg's invention in Germany in 1930s
 - Patented in Germany in 1936
 - About 50 years ahead of its time

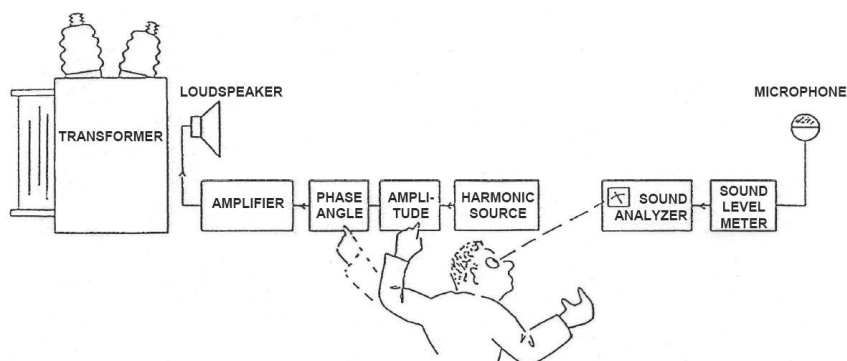


Ref: P. A. Nelson and S. J. Elliott, "Active Control of Sound," 1992.

5

Manual ANC of the 1950s

- Conover's manual noise control system for a transformer



Ref: P. A. Nelson and S. J. Elliott, "Active Control of Sound," 1992.

6

Audio Examples

Vacuum cleaner



Transformer



Audio examples taken from: David Crawford, 1996,
<http://www.spd.eee.strath.ac.uk/~david>

7

Practical applications of ANC

- Ventilation systems in offices and in industry
- Active hearing protectors and headphones
 - Combination of passive and active noise control gives the best result
- Vehicle cabins in cars, trains, aircrafts
 - Passenger comfort!
- Snoring suppression systems
 - At least 2 commercial systems developed in the US (1992, 2004)
(Ref. Personal communication with Prof. Scott C. Douglas, Southern Methodist Univ., Dallas, TX)
- In most applications, the ANC system must be adaptive
 - The acoustic transfer function is affected by changes in temperature, humidity, dirt, flow, speakers, microphones

8

Experimental Snore ANC System



Picture taken from Kajikawa et al. 2012

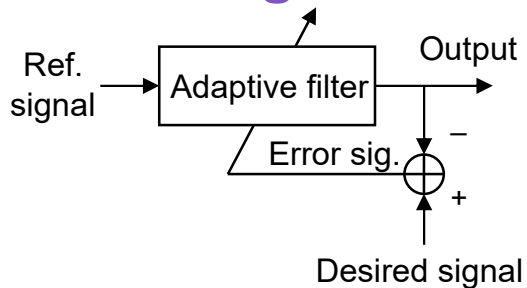
Introduction of Adaptive Filtering

- Coping in a changing environment requires adaptation
 - General principle in nature: adapt or die
- An adaptive filter is necessary, when the desired response of the filter varies over time
- Widrow and Hoff discovered **LMS algorithm** in 1959
 - LMS = least mean squares
- Related to the research that also led to the development of ADALINE, a neural network for pattern recognition



Basic Idea of Adaptive Filtering

repeat forever
 Filter one sample
 compute error
 Adjust parameters
 end



- Must know the desired signal to define the error
- In ANC: desired signal = original noise
- A rule is needed to adjust the filter parameters to reduce the error
- After converge, the error signal should be close to zero (“silence”)

11

LMS Algorithm

- Rule to adjust FIR filter coefficients, minimizing the least squares error (Widrow-Hoff algorithm):

$$\mathbf{w}(n+1) = \mathbf{w}(n) + 2\mu\mathbf{e}(n)\mathbf{x}(n)$$

where $\mathbf{w}(n)$ is the filter parameter vector,

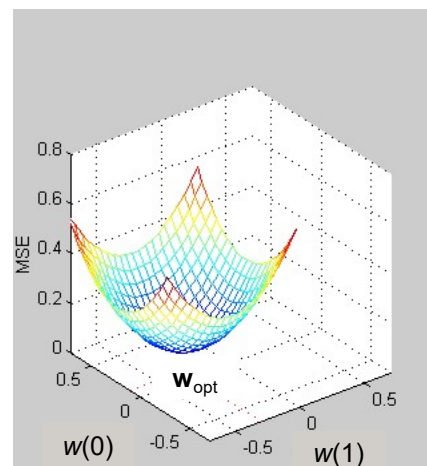
μ is an adaptation constant

$\mathbf{x}(n)$ is the reference (input) signal vector

$\mathbf{e}(n) = \mathbf{d}(n) - \mathbf{y}(n) = \mathbf{d}(n) - \mathbf{w}^T(n)\mathbf{x}(n)$ is the error signal, and

$\mathbf{d}(n)$ is the desired signal

- Initial values of \mathbf{w} don't matter
- Constant μ must not be too large or too small



12

Pros and Cons of LMS Algorithm

Pros

- Simple and effective
- Small memory footprint (delay lines and parameters)
- Still a very popular algorithm
- Can be modified in various ways (frequency-domain version, sign-only version, apply filters in signals, among others)

Cons

- Slow convergence
- Convergence speed depends on signal (power, spectral shape)
- Too "heavy" for some low-cost applications or when the impulse response is very long

13

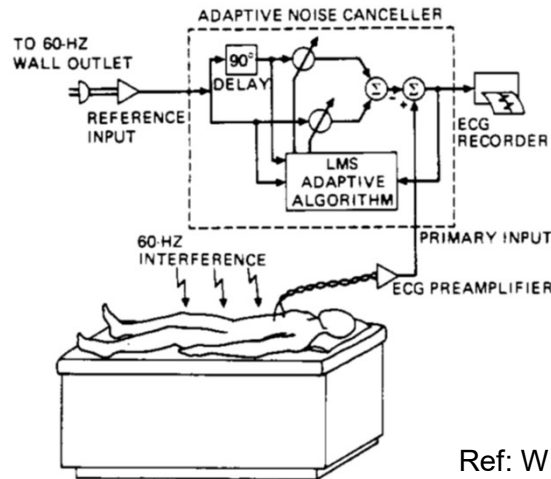
Adaptive Interference Cancellation

- Removal of disturbances from a digitized electronic signal
- An early application: removal of 60-Hz power grid disturbance from ECG measurements (electrocardiogram, heartbeat signal) (Widrow et al., 1975)
- Removal of background noise in communications systems
 - For example, smartphones, airline pilots...
- Removal of periodic disturbance without reference signal
 - If the frequencies are known, a synthetic reference signal can be generated
 - For example, removal of 50/60-Hz power-line hum from a recording
 - A delayed reference signal can be used, if it remains stationary
 - Special case: DC removal



14

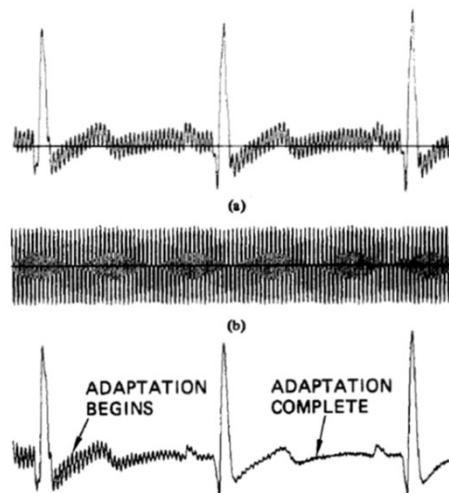
Adaptive Interference Cancellation



Ref: Widrow *et al.*, 1975

15

Adaptive Interference Cancellation

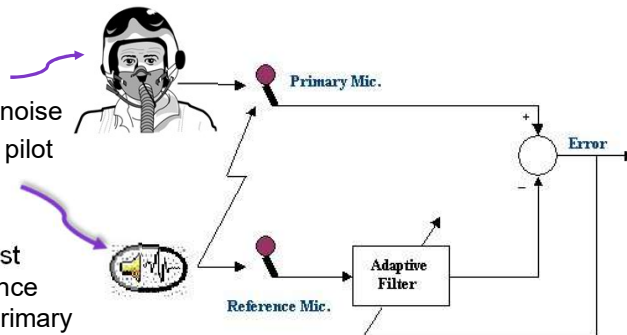


Ref: Widrow *et al.*, 1975

16

Noise Removal from Helmet Mic Signal

- Two mics are needed
 1. Primary mic inside the helmet capturing the speech signal plus noise
 2. Reference mic far away from the pilot to capture the ambient noise
- The adaptive filter will find the best parameters to remove the reference signal (ambient noise) from the primary mic signal
- The error signal is sent to the receiver (less noisy than the primary mic signal)



Ref: <http://www.eas.asu.edu/~dsp/grad/anand/java/ANC/ANCDesc.html>

17

Interference Cancellation Demo

- A guitar player thinks that his recording is damaged, but an adaptive filter can fix it!
 - A reference signal is needed, but it is easy to acquire
- 🔊 Corrupted recording (“La Foa”)
- 🔊 Interference only (50 Hz + harmonics)
- 🔊 After adaptive processing ($N = 9$; $\mu = 0,005$)
- 🔊 After adaptive processing ($N = 9$; $\mu = 0,1 \dots 0,001$)
 - First fast convergence, then slow
- 🔊 Original recording with interference (“La Foa”)

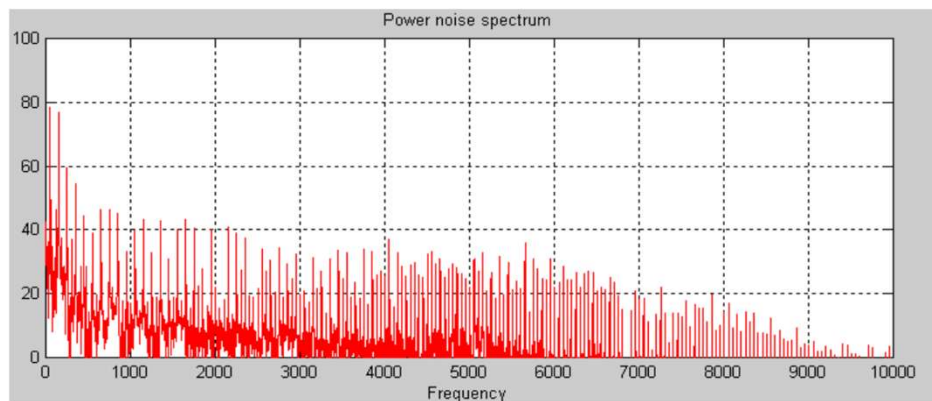


Audio demo by Tommi Huutilainen and N. N. (2001)

18

Power Line Interference

- 50 Hz and its odd harmonics
- Wideband signal containing many peaks



19

Duct ANC Project in Finland

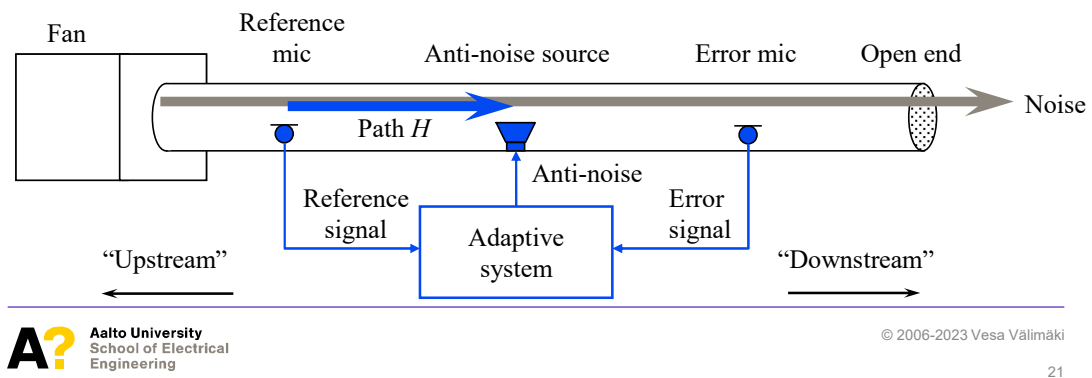
- Particularly well suited application for ANC
 - Fan noise propagates along the duct without attenuation
 - Passive attenuators are large and resist flow
 - 1-D case: ANC is "easy" below the frequency limit of transversal propagation



20

Duct ANC: Basic System

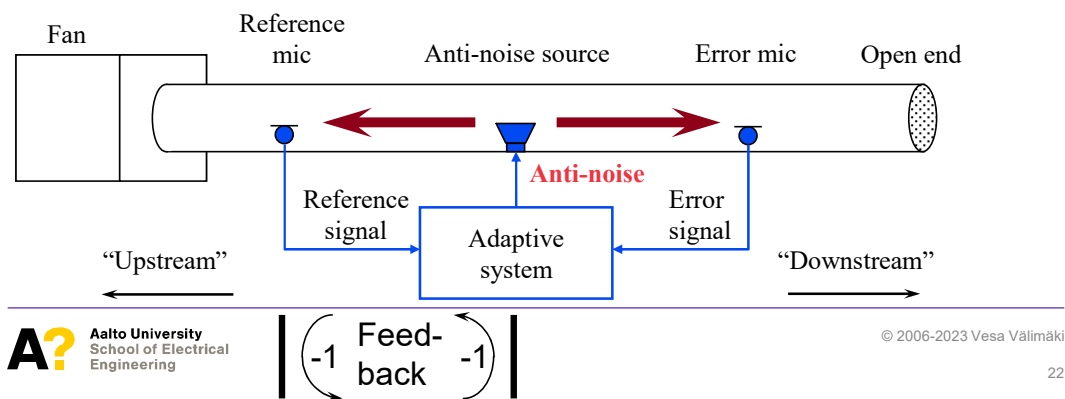
- Adaptive system produces the anti-noise by filtering a reference signal
 - Reference mic captures the noise: Reference signal
 - Adaptive system models the acoustic propagation path H and inverts the signal
 - Error mic is used for checking whether the suppression succeeds
- However, problems arise using omnidirectional mics and loudspeakers!!!



21

Acoustic Feedback in ANC

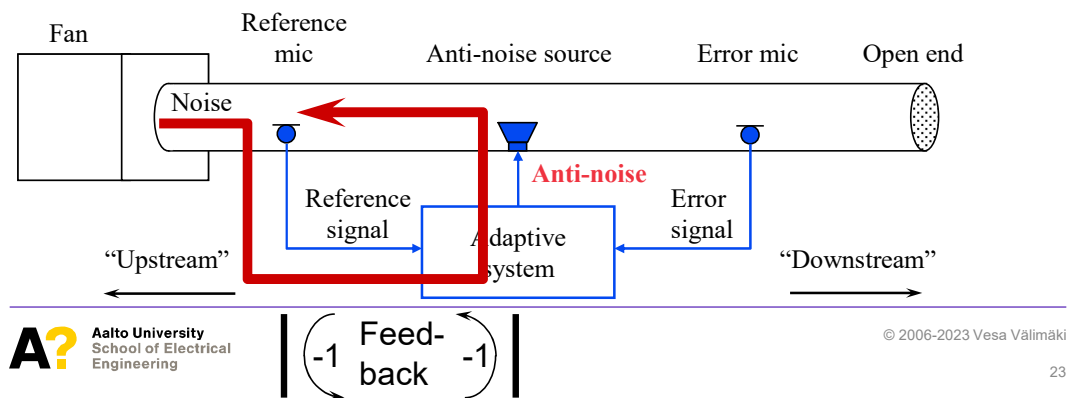
- A regular speaker radiates sound in both directions
- A regular mic hears equally well from both directions
 - Anti-noise gets mixed with the reference mic signal, reducing attenuation
- A high risk of **howling** caused by acoustic feedback!!!



22

Problem from Upstream Direction

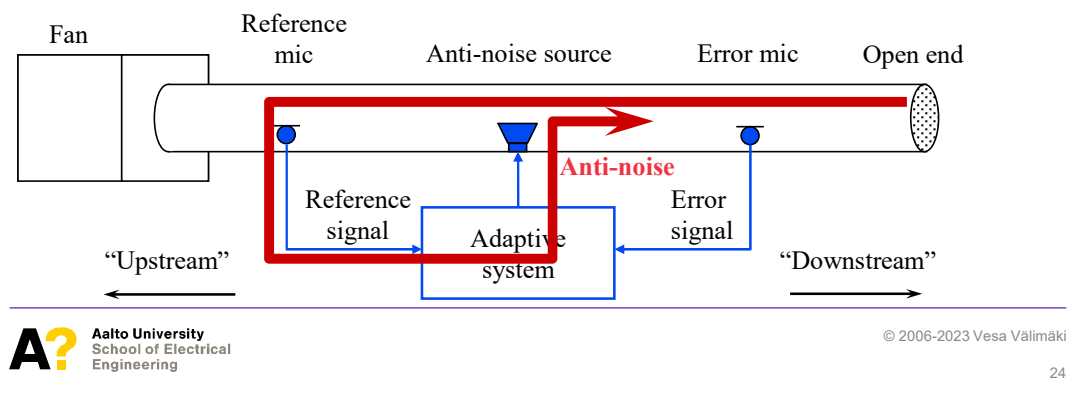
- The anti-noise speaker acts as a reflector
 - New standing waves are formed inside the duct because of the active system
 - Noise level is increased on the upstream side
 - The system can start howling (acoustic feedback)



23

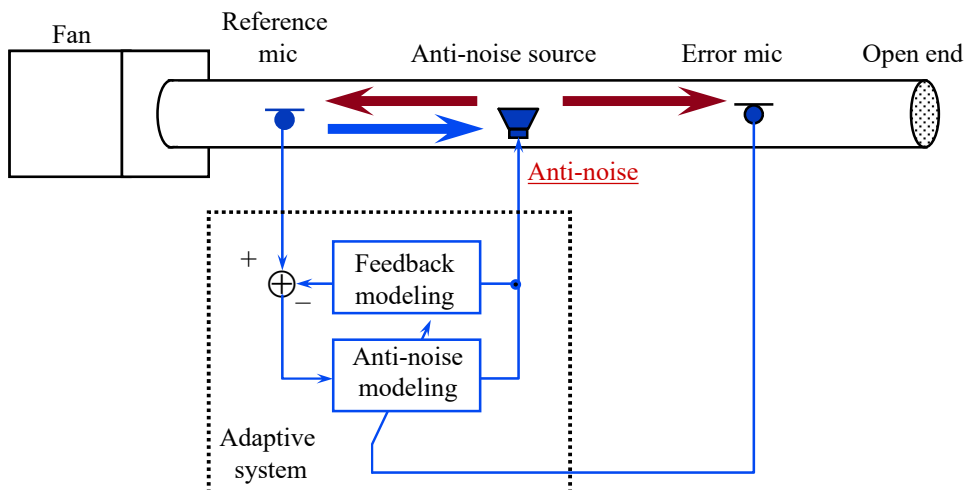
Problem from Downstream Direction

- The reference mic acts as a reflector
 - New standing waves are formed between the ref mic and the open end of the duct
 - Acoustic properties of the duct are changed and noise level may increase



24

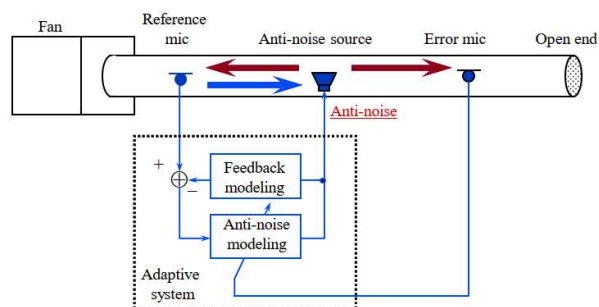
A Solution: Feedback Cancellation



25

A Solution: Feedback Cancellation

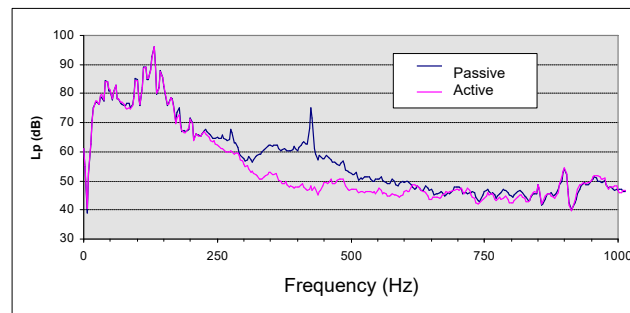
- The feedback is modeled with another adaptive filter and is cancelled
 - The same idea as “echo cancellation” in telephone systems
- The anti-noise system will not howl so easily 😊
- The ANC speaker still radiates in both directions ☹️



26

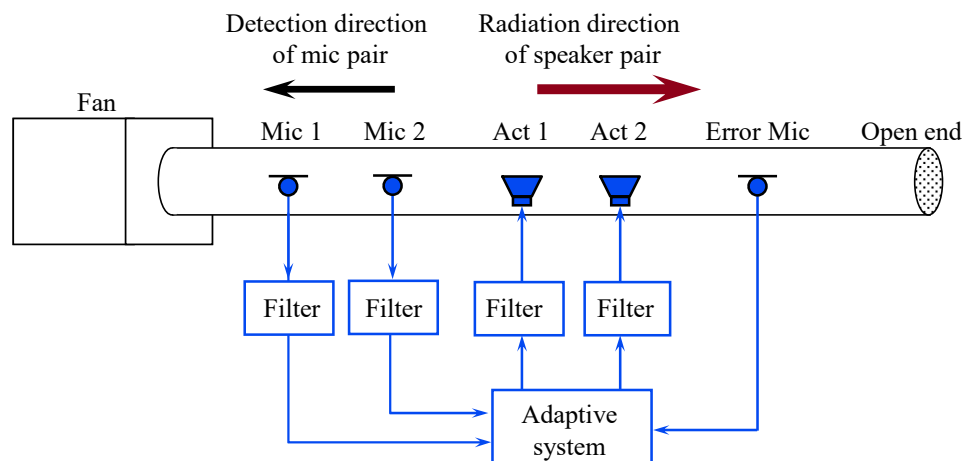
Results with Feedback Cancellation

- A loudspeaker was used as the noise source
 - Air flow disturbs the ANC system
- Near the blade-passing frequency, attenuation was achieved on a wide band
 - Over 20 dB of attenuation at BPF, but only a few dB in average



27

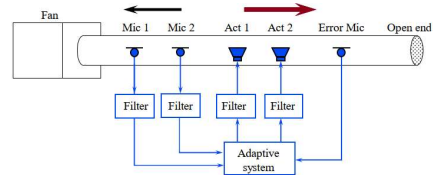
More Advanced Solution: Unidirectional Mics and Speaker



28

Advantages of a Unidirectional ANC System

- The anti-noise does not radiate upstream ☺
 - Noise level does not increase in the wrong place
- The unidirectional mic does not hear from downstream ☺
- Acoustic feedback is effectively suppressed! ☺
- Also isolates the anti-noise signal from the reference signal, improving the performance of the ANC system

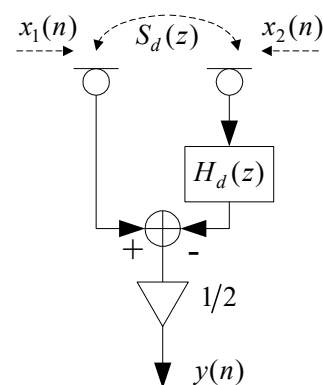


(Ref. Seppo Uosukainen, Doctoral dissertation, Helsinki Univ. of Tech., 1999)

29

Principle of the Unidirectional Mic Pair

- Originally developed for a speaker pair, but it is reversible
- The signal of one sensor is delayed appropriately
 - Cancels signal x_2 coming from downstream
 - Side effect: Signal x_1 coming from upstream gets highpass filtered
- The delay between sensors must be very accurate
 - Need a fractional delay filter or a measured impulse response between the sensors
- Assumption: Transfer function $S_d(z)$ is pure delay
 - Losses or reflections are not accounted for



30

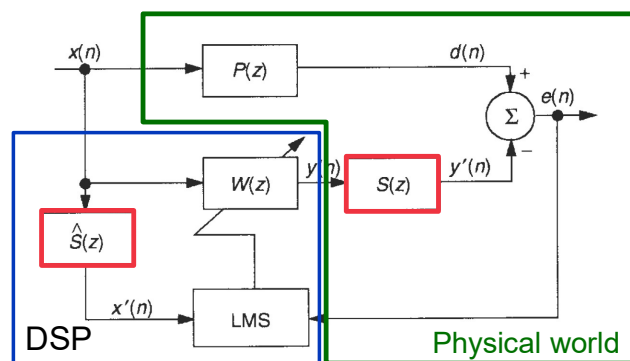
FXLMS Algorithm

- In many acoustic ANC cases, the error signal cannot be measured directly
 - Error mic is not at the same location as the anti-noise source (speaker)
- However, the LMS algorithm does not converge, if the error signal is not synchronized with the reference signal
- The FXLMS algorithm (Filtered-X LMS) uses a filter estimate $S(z)$ for the error path
 - Reference signal $x(n)$ is filtered with $S(z)$ to compensate for the delay between the error and reference signals
- This algorithm was discovered by several researchers at the same time (Morgan, 1981; Burgess, 1981; Widrow 1981)

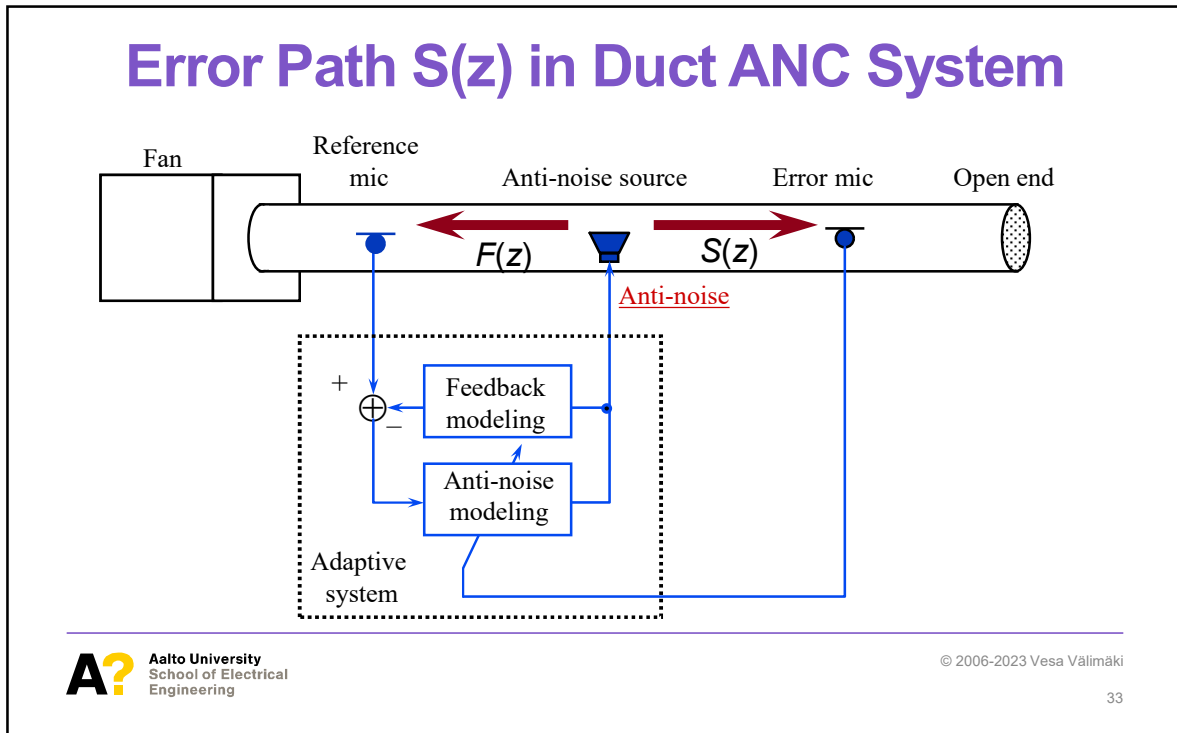
31

FXLMS Algorithm

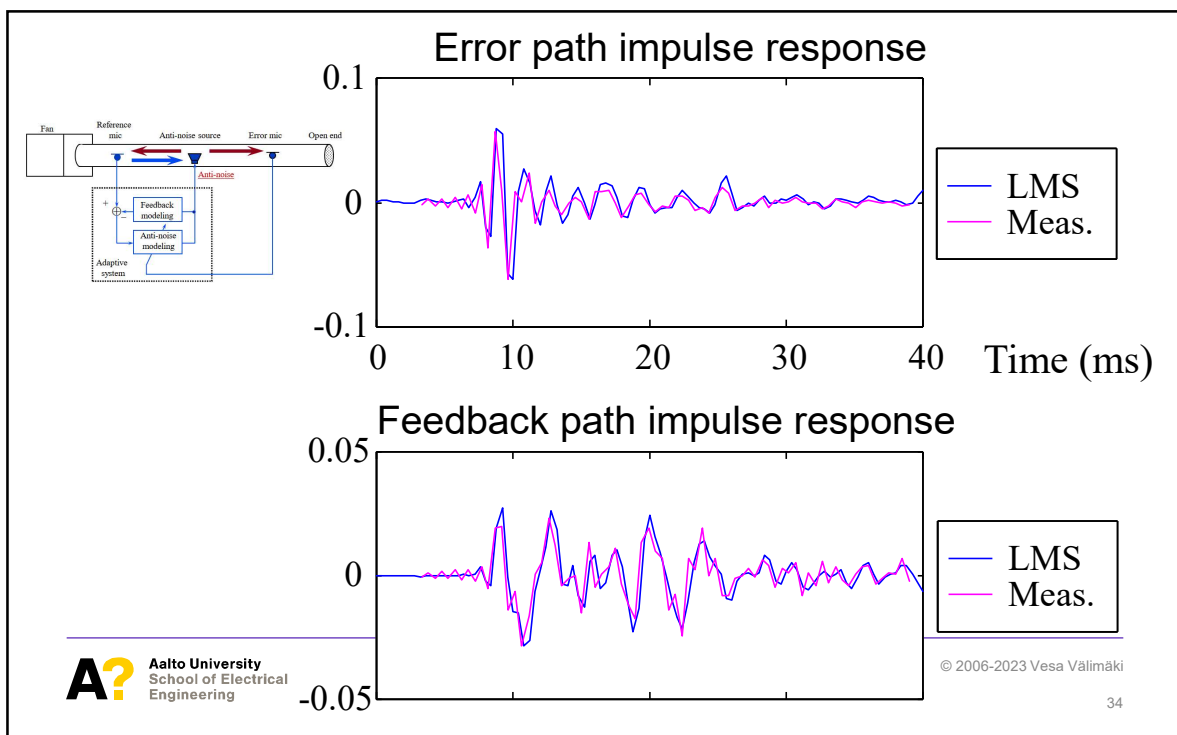
- FXLMS algorithm (Filtered-X LMS) uses a filter $S(z)$ to approximate how the error signal is delayed and processes the reference signal in the same way (Morgan, 1981; Burgess, 1981; Widrow 1981)



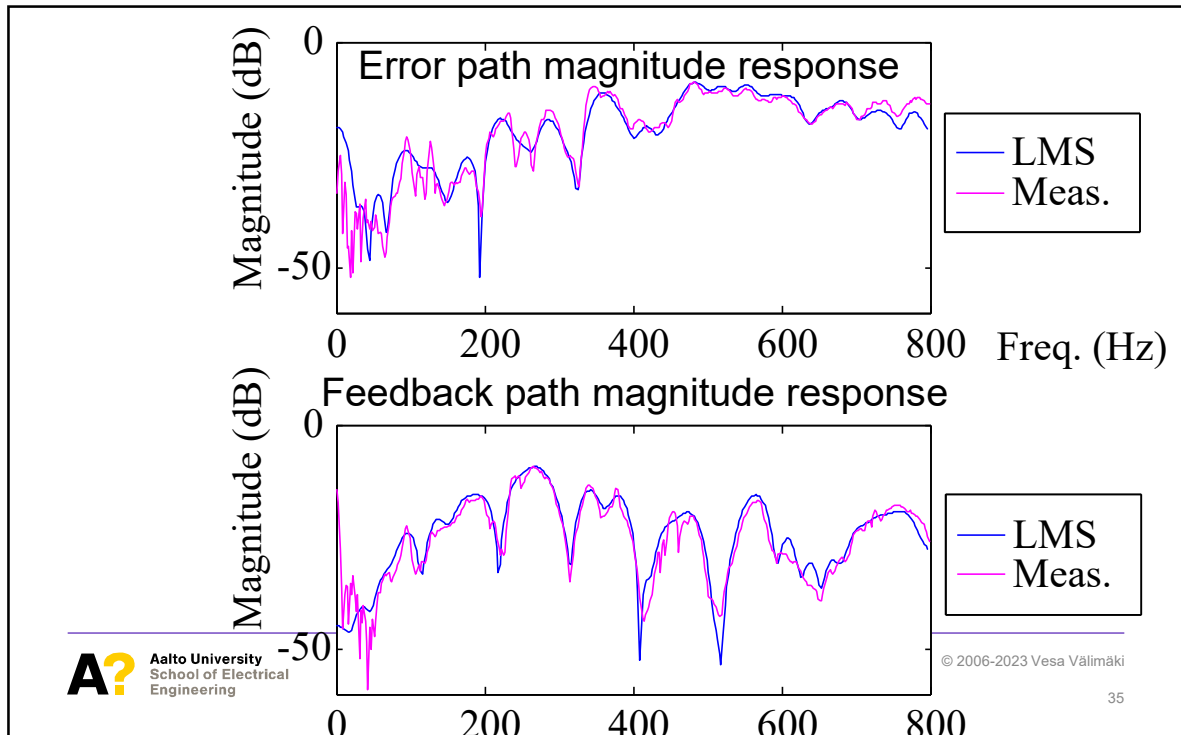
32



33



34

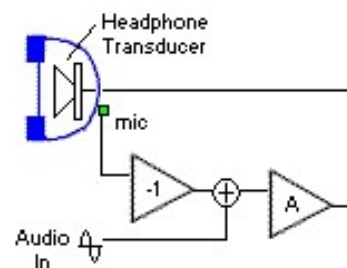


35

ANC Headphones

- A low-cost **analog** ANC headphone registers the ambient noise with a mic and uses an inverting amplifier to play it as anti-noise
 - Gain is tuned at factory (non-adaptive)
- Reduces low-frequency ambient noise about 10 to 20 dB, but only on a narrow band
 - Passive materials help at high frequencies
- Audio signal can be played at the same time, without interaction with the ANC function
- Audio demo by Jussi Rämö (Aalto Univ. 2011): *Bose Quiet Comfort 2*

Ambient noise – Bose OFF – Bose ON

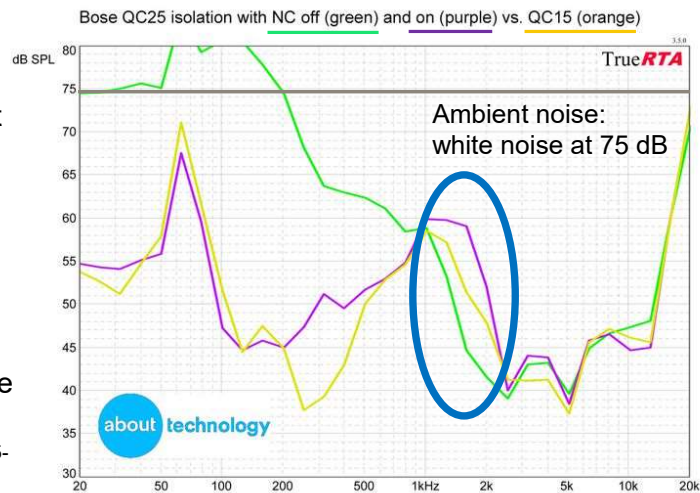


36

ANC Headphones

- ANC attenuates noise most effectively at low frequencies below about 200 Hz
- At higher frequencies, the wavelength of sound limits its functioning
- ANC can boost noise near its cutoff frequency (“**waterbed effect**”), here between 1 and 2 kHz

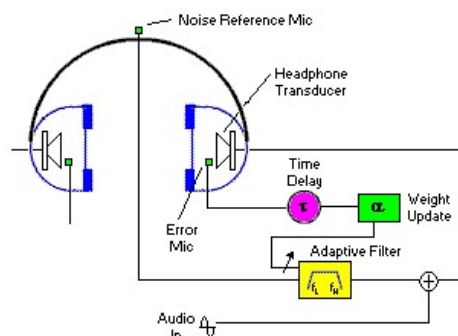
Ref: <https://www.lifewire.com/bose-qc25-review-specs-3134560>



37

Digital ANC Headphones

- A digital feedforward ANC headphone requires a **reference mic** outside the headphones
 - Often there are several reference mics, to capture ambient noise from various directions
 - Another external mic or mic pair may be used to capture the user's own voice for smartphone use
- An **error mic** is needed inside each earpiece for adaptation



38

ANC in Airplanes



- The cabin of a modern turboprop airplane is fairly noisy
 - The blade-passing noise occurs at low frequencies and propagate easily through the fuselage
- Airplanes must be light, so passive attenuation of low-freq noise is hard
- ANC seems like a nice solution
 - ANC works best at low frequencies
 - ANC system does not add much to the weight of the aircraft
 - The cost of an ANC system is reasonably small w.r.t. to the price of the aircraft

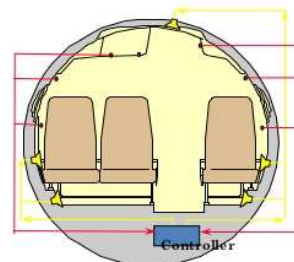
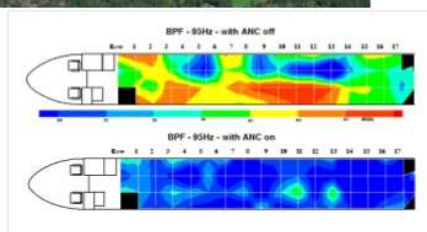
39

ANC in Airplanes

Saab 2000 and Gripen



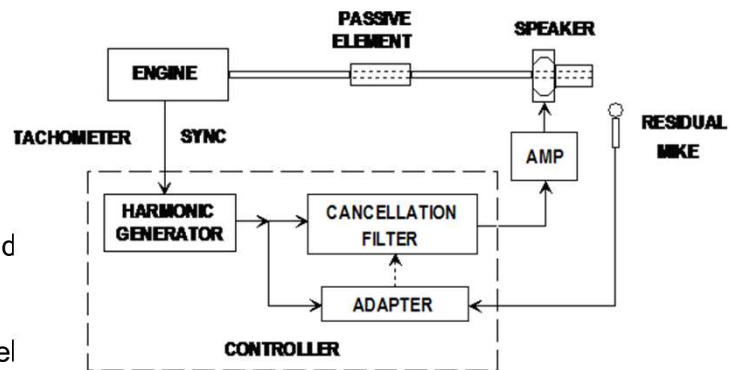
A 400M



40

Active Muffler for Cars

- An ANC muffler uses synthetic engine harmonics to produce anti-noise
- A regular muffler reduces the engine power
- An active muffler would not reduce power that much
- Could it reduce the fuel consumption?



Ref: <http://doctord.dyndns.org/Pubs/POTENT.htm>

41

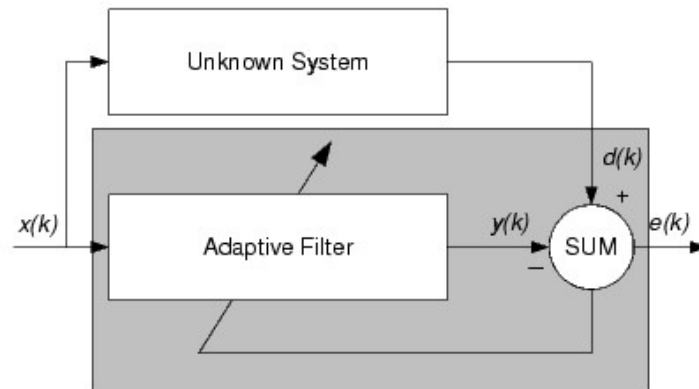
Use Cases of Adaptive Filtering

- 1) Transfer function modeling
 - Adaptive filter learns a (time-varying) frequency response
 - For example, active noise control
- 2) Inverse modeling
 - Adaptive filter cancels of transfer function
 - For example, automatic equalization of a loudspeaker
- 3) Prediction
 - Adaptive filter learns to "predict" the output signal of a system
 - For example, predictive coding, ADPCM, spectral envelope models
- 4) Interference cancellation
 - Adaptive filter "subtracts" an interference from a corrupted signal

42

Transfer function modeling

- Adapt the filter so that the filter responses become equal (i.e. error signal becomes zero)

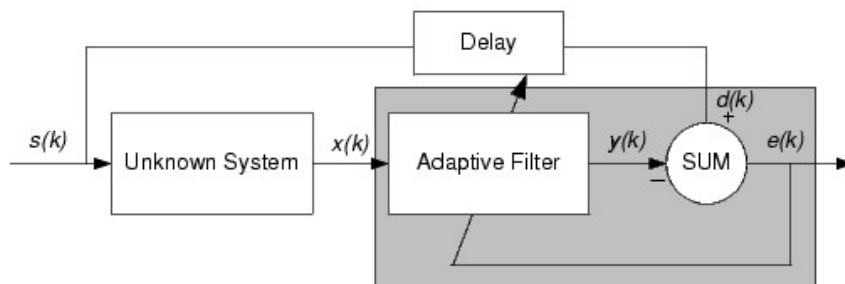


Ref: <http://www.mathworks.com/access/helpdesk/help/toolbox/filterdesign/adaptiv4.shtml>

43

Inverse modeling

- Adaptive filter learns to cancel the response of a system
- A **modeling delay** must be included!

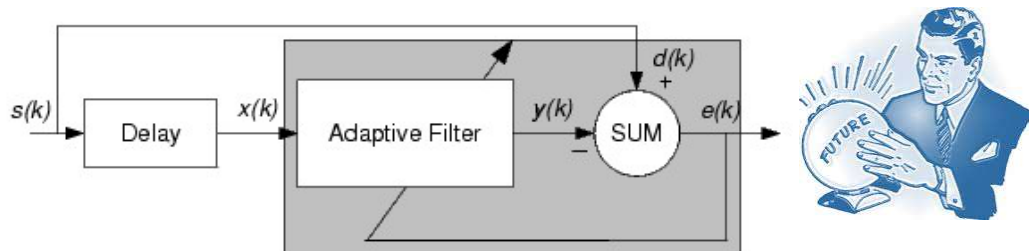


Ref: <http://www.mathworks.com/access/helpdesk/help/toolbox/filterdesign/adaptiv5.shtml>

44

Prediction

- Adaptive filter learns to make the best guess of next signal value
- Learning correlation properties is equivalent to spectral estimation

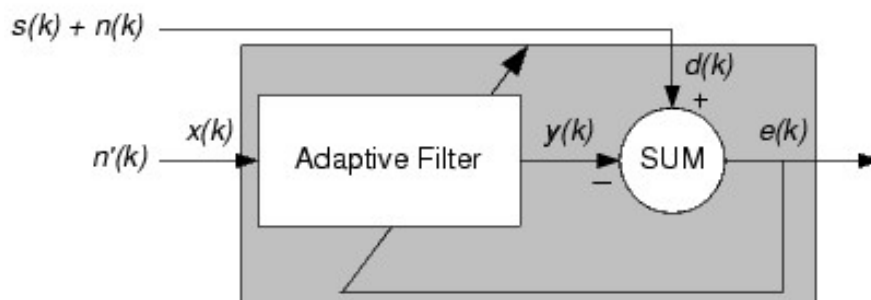


Ref: <http://www.mathworks.com/access/helpdesk/help/toolbox/filterdesign/adaptiv7.shtml>

45

Interference Canceling

- Subtract the filtered interfering signal $n'(k)$ from the mixture of desired signal and interference $s(k) + n(k)$



Ref: <http://www.mathworks.com/access/helpdesk/help/toolbox/filterdesign/adaptiv6.shtml>

46

Adaptive Filtering Applications

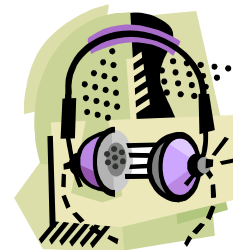
- Automatic equalization of audio system
 - E.g. home theater systems
- Echo cancellation
 - Telephone systems (since 1965)
 - Hands-free communication systems
 - Teleconference systems
- Feedback cancellation in hearing aids
- Noise equalization
 - In working machines, it is necessary to reduce noise, but it must not disappear
 - In car cabins, the engine noise can be made more sporty
- Beam forming
 - E.g. mic arrays in teleconferencing



47

Conclusion

- Active noise control is an old idea (1930s...)
- Used in combination with passive noise control
 - Active control work well at low frequencies, passive attenuation work best at high frequencies
- Active hearing protectors and headphones are the most successful application so far
- In the future, there may be more ANC applications
 - Many devices can have enough computing power, mics, and speakers
 - ANC can bring increased **comfort or savings**
 - Environmental noise is becoming a major issue but there are still not many ANC applications (Lam *et al.*, 2021)
 - Patents are expiring (US patents are valid for 20 years)



48

References

- J. C. Burgess, "Active adaptive sound control in a duct—A computer-simulation," *J. Acoust. Soc. Am.*, vol. 70, pp. 715–726, 1981.
- Y. Kajikawa, W.-S. Gan, and S. M. Kuo, "Recent advances on active noise control: Open issues and innovative applications," *APSIPA Transactions on Signal and Information Processing*, vol. 1, pp. 1–21, 2012.
- S. M. Kuo and D. R. Morgan, "Active noise control: a tutorial review," *Proc. IEEE*, vol. 87, no. 6, pp. 943–973, June 1999.
- S. M. Kuo and D. R. Morgan, *Active Noise Control Systems: Algorithms and DSP Implementations*. Wiley, 1996.
- B. Lam *et al.*, "Ten questions concerning active noise control in the built environment," *Building and Environment*, vol. 200, paper no. 107928, May 2021.
- D. R. Morgan, "An analysis of multiple correlation cancellation loops with a filter in the auxiliary path," *IEEE Trans. Acoust. Speech Signal Process.*, vol. 28, pp. 454–467, Aug. 1980.
- P. A. Nelson and S. J. Elliott, *Active Control of Sound*. Academic Press, 1992.
- B. Widrow *et al.*, "Adaptive noise cancelling: principles and applications," *Proc. IEEE*, vol. 63, no. 12, pp. 1692–1716, Dec. 1975.
- B. Widrow, D. Shur, and S. Shaffer, "On adaptive inverse control," in *Proc. 15th Asilomar Conf. Circuits Systems and Computers*, pp. 185–189, 1981.
- V. Välimäki, J. Kataja, and M. Antila, "Unidirectional solutions for active noise control in ducts," in *Proc. INTERNOISE'2001*, pp. 635–640, The Hague, The Netherlands, Aug. 2001.

References

Readings for the exam

- S. J. Elliott, "Down with noise," *IEEE Spectrum*, pp. 54–61, June 1999.
- B. Rafaely, "Active noise reducing headset: an overview," in *Proc. INTERNOISE'2001*, The Hague, The Netherlands, Aug. 2001.