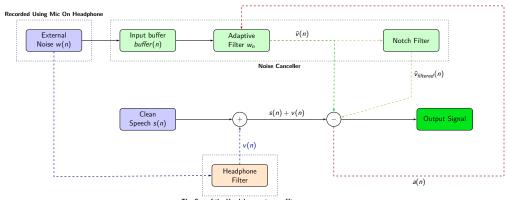
# Block Diagram

#### **Adaptive Noise Cancellation System**

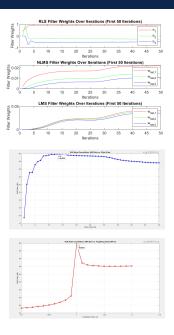


The Cup of the Headphone acts as a filter



## Design Choices

- We chose to use RLS rather than LMS or NLMS because RLS converges faster than LMS and NLMS.
- The RLS algorithm is more complex than LMS and NLMS, but it provides better performance in terms of convergence speed and steady-state error. We can see that from the plot filter coefficients v/s iterations
- The optimal value of the filter order p is determined by SNR v/s filter order plot for lambda = 1. The optimal value of p obtained is 13.
- ullet The optimal value of the forgetting factor lambda is determined by SNR v/s forgetting factor plot for p = 13. The optimal value of lambda obtained is 1.



### Noise Canceller Design

 The RLS algorithm follows the following Error Function:

$$E(n) = \sum_{i=0}^{n} \lambda^{n-i} e^{2}(i)$$

$$e(i) = d(i) - w_{n}^{T} \times (i)$$

 After taking the derivative w.r.t w<sub>n</sub> and setting it to zero and simplifying it, we get the following solution for the weights:

$$R_X(n)w_n = r_{dX}(n)$$

• We now try to calculate  $w_{n-1}$  using the above equation by calculating  $R_X^{-1}(n)(P(n))$ . We finally get the update equation for the weights as:

$$w_n = w_{n-1} + g(n)\alpha(n)$$

Here.

$$g(n) = \frac{P(n)x(n)}{\lambda + x^{T}(n)P(n)x(n)}$$
$$\alpha(n) = e(n) - w_{n-1}^{T}x(n)$$

$$H(z) = \frac{1 - 2\cos(\omega_0)z^{-1} + z^{-2}}{1 - 2r\cos(\omega_0)z^{-1} + rz^{-2}}$$

$$\Rightarrow \frac{Y(z)}{X(z)} = \frac{1 - 2\cos(\omega_0)z^{-1} + z^{-2}}{1 - 2r\cos(\omega_0)z^{-1} + rz^{-2}}$$

$$\Rightarrow y(n) = a_1y(n-1) - a_2y(n-2) + b_0x(n) - b_1x(n-1) + b_2x(n-2)$$

- The values of a<sub>1</sub>, a<sub>2</sub>, b<sub>0</sub>, b<sub>1</sub>, and b<sub>2</sub> are initialized based on the transfer function.
- The noise may contain multiple tonal frequencies, so we have cascaded multiple notch filters to suppress all the tonal frequencies.
- The value of r is set very close to 1 to reduce the width at which the falloff occurs. The more we increase the value of r, the narrower the notch filter becomes.

- Pros-
- RLS converges much faster than LMS and NLMS.
- It works better in the real world where the noise is not stationary. LMS and NLMS are not able to adapt to the changing noise environment as quickly as RLS.
- The RLS algorithm provides better SNR and MSE performance than LMS and NLMS.
- The Notch filter is better than trying to use a bandpass filter using the FilterDesigner tool in MATLAB as it is able to suppress the tonal frequencies much faster than a bandpass filter.
- Cons:
- The RLS algorithm is more complex than LMS and NLMS, which makes it harder to implement.
- The RLS algorithm requires more computational resources than LMS and NLMS, which can be a problem for real-time applications.
- For stationary noise, with some non-stationary components, the RLS algorithm performs similarily to LMS and NLMS.

#### References



J. G. Proakis and D. G. Manolakis, *Digital Signal Processing: Principles, Algorithms, and Applications*, 4th ed., Prentice Hall, 2007.



M. H. Hayes, Statistical Digital Signal Processing and Modeling, John Wiley & Sons, 1996.