# An active noise control casing using the multi-channel feedforward control system and the relative path based virtual sensing method

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## Abstract

Active noise control (ANC) is a noise reduction technique based on acoustic wave superposition. The sound pressure level is reduced in a zone of quiet (ZoQ) by an anti-noise wave transmitted from the control source. The antinoise wave has the same amplitude and opposed phase of the noise wave. An error microphone is conventionally placed at the target ZoQ to monitor the sound pressure level, forming a closed-loop control in an ANC system. However, due to application constraints or physical limitations, the error microphone sometimes cannot be placed at the target ZoQ. Virtual sensing (VS) methods are developed for such situations. There are two most commonly used VS methods. They are the auxiliary filter based VS (AF-VS) method and the remote microphone based VS (RM-VS) method. The AF-VS method preserves the information regarding the optimal control filter that can achieve the maximum noise reduction at the target ZoQ based on remote measurements. In this paper, we propose a new VS method, the relative path

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based VS (RP-VS) method, which estimates both the disturbance signal and the anti-noise signal at the target ZoQ. A theoretical analysis is provided to demonstrate that under different assumptions of varying acoustic paths, the RP-VS method can behave in the same way as the AF-VS method or the RM-VS method. Simulation results validate this theoretical analysis and demonstrate that improved noise reduction can be achieved by the RP-VS method when the noise frequency varies. Lastly, an ANC casing is built up with the RP-VS method to reduce a varying broadband fan noise. The RP-VS method is validated to be as effective as the AF-VS method and the RM-VS method with the implementation of the ANC casing.

*Keywords:* Active noise control, multi-channel feedforward control, virtual sensing

## 1. Introduction

Noise pollution is a pressing environmental concern in our modern times. There are two fundamental ways to abate noise, passively and actively. The passive noise control (PNC) absorbs and diffracts the noise wave, and therefore can effectively deal with high-frequency noise. However, the efficiency of PNC decreases when the noise frequency is relatively low, due to the massive size, high cost and complexity in deployment [1, 2]. The active noise control (ANC) uses acoustic actuators such as loudspeakers to emit an anti-noise wave. When the anti-noise wave has the same amplitude and opposed phase as the unwanted noise wave, the sound pressure level can be reduced in a zone of quiet (ZoQ) based on the acoustic wave superposition [3]. ANC is



Active Noise Control Casing

Figure 1: Illustration of the ANC casing.

ideally suited to reduce low-frequency noise and is therefore an irreplaceable complement to PNC.

Besides its great implementation success in noise canceling headphones, ANC has recently provided a source of hope due to its possibility in reducing noise in smart cities, buildings, manufacturing and transportation [4, 5]. An ANC casing is proposed to enclose a noise source inside a sound-proof shield with an opening to allow heat and air ventilation [6]. As illustrated in Fig. 1, control sources are distributed at the opening to transmit the anti-noise wave that cancels the noise wave emitted by the enclosed noise source. A small-scale ANC casing may be integrated with household appliances and home servers, while a large-scale ANC casing is possible to be adapted for machinery such as electrical transformers, padding machines, and so on.

Being an ANC application, ANC casings adopt acoustic sensors such

as microphones to provide real-time information of the noise wave for the calculation of control signals. According to microphones, ANC systems are categorized into feedforward and feedback systems. The feedforward ANC system contains both reference and error microphones, while the feedback ANC system consists of only error microphones. The reference microphones measure the noise wave upstream, in order for the ANC controller to be fed with reference signals that are highly correlated with error signals that are acquired by the error microphones. Hence, the feedforward ANC system is more efficient in reducing the broadband noise. In the feedback ANC system, the reference signal is estimated by internal models based on error signals. Usually, such estimation is only effective in reducing the narrowband noise.

ANC systems are also categorized by the number of loudspeakers used. A single-channel ANC system consists of one loudspeaker, often together with one error microphone and at most one reference microphone. A multi-channel ANC system includes multiple loudspeakers and several microphones [7, 8]. It can form a relatively large ZoQ as compared to the noise wavelength. The ANC casing is designed to be a multi-channel feedforward ANC (MCFFANC) system.

Adaptive ANC controllers use error signals to update their control filter coefficients. The filtered-x least mean squares (FxLMS) is the most widely used adaptive algorithm in ANC systems [9]. The goal of the FxLMS algorithm is to minimize 2-norm of error signal vector, which includes instant samples from all the error microphones [10]. This ensures that the ZoQ is formed around error microphones [11]. In the design of an ANC casing, error microphones have to be placed close enough to control sources to avoid obvious protuberance. Therefore, the FxLMS algorithm can only achieve local noise reduction. For better global noise reduction performance, the ANC casing desires the ZoQ to be formed far away from the error microphones.

Virtual sensing (VS) methods are developed to solve this dilemma [12, 13, 14, 15]. Currently, there are two virtual sensing methods that are commonly used in implementations of ANC systems. They are the auxiliary filter based VS (AF-VS) method and the remote microphone based VS (RM-VS) method. The AF-VS method preserves the information about the optimal noise control filter that can achieve the maximum noise reduction at the target ZoQ [16, 17, 18]. The RM-VS method estimates the disturbance signal at the target ZoQ based on the measurable error signals [19, 20]. The AF-VS method and the RM-VS method are recently compared by different experimental configurations of control sources and error microphones [21, 22]. A delayed RM-VS method is thus proposed to overcome the revealed causality weakness of the RM-VS method [23, 24]. However, there are two practical aspects that have yet to be thoroughly investigated. They are the varying acoustic paths and varying noise characteristics [25, 26]. A successful VS method should be robust to changes in acoustic paths and noise characteristics.

The rest of this paper is organized as follows. Firstly, the AF-VS method and the RM-VS method are reviewed and the relative path based VS (RP-VS) method is proposed. The RP-VS method estimates not only the disturbance signal but also the anti-noise signal at the target ZoQ. It can behave in the same way as the AF-VS method or the RM-VS method, under different assumptions of varying acoustic paths. Secondly, the aforementioned three VS methods are compared with the fixed-coefficient (FC) filter. The FC filter is the most commonly used non-adaptive method in commercial ANC applications, due to its effectiveness and simplicity. Varying acoustic paths measured from experimental setups of the single-channel and dual-channel feedforward ANC systems are adopted. The change of noise characteristics is carried out by setting different noise frequency bands. The theoretical analysis and simulations results validate that the proposed RP-VS method is an effective VS method and robust to the demonstrated changes in acoustic paths and noise frequency bands. Lastly, an ANC casing is constructed. The experiment results show that when the speed of a computer fan is accelerated, the noise reduction performance achieved by the RP-VS method is as good as the AF-VS method and the RM-VS method at the target ZoQ.

#### 2. Virtual Sensing Methods

There are two stages when the VS method is in use. The first stage is the tuning stage. In this stage, temporal microphones, which are referred to as the virtual microphones, can be placed at the target ZoQ in order to train the control filter and model the transfer functions between virtual microphones and error microphones. The error microphones are placed far from the target ZoQ. They are also referred to as the monitoring microphones. The second stage of the VS method is the control stage. Without any microphones placed at the target ZoQ, adaptive control filters converge based on real-time outputs of the monitoring microphones and the prior information obtained in the tuning stage.



(a) Tuning stage



Figure 2: Block diagram of the AF-VS method.

#### 2.1. Auxiliary filter based virtual sensing method

Figure 2 shows the block diagram of the AF-VS method in z domain. Notations of signals and acoustic paths are different in the tuning stage and the control stage. In this way, we are able to analyze the effect of varying acoustics paths and varying noise characteristics on the converged control filter and corresponding noise reduction. These effects are overlooked in previous works on VS methods.

In the tuning stage, the control filter is firstly converged to the optimal solution  $W_o(z)$  that minimizes the power of the virtual error signal  $E_v(z)$ . The virtual error signal  $E_v(z)$  is provided by the temporal microphone placed at the target ZoQ, which is written as the acoustic superposition of the noise wave and the anti-noise wave, *i.e.* 

$$E_{v}(z) = D_{v}(z) + S_{v}(z)Y(z).$$
(1)

In Eq. (1),  $D_v(z) = P_v(z) X(z)$  is the disturbance signal.  $Y(z) = W_o(z) X(z)$  is the control signal. The virtual primary path  $P_v(z)$  and virtual secondary path  $S_v(z)$  are the transfer functions from the noise source and control source to the virtual microphone, respectively. X(z) is the reference signal.

The yellow block indicates a perfect model of the virtual secondary path that is assumed to be available for the controller in the tuning stage. The zdomain optimal solution yields that

$$W_{o}(z) = -\frac{P_{v}(z) X(z)}{S_{v}(z) X(z)} = -\frac{P_{v}(z)}{S_{v}(z)} B(X), \qquad (2)$$

where

$$B(X) = \begin{cases} 1 & X(z) \neq 0 \\ 0 & X(z) = 0 \end{cases}$$
(3)

Meanwhile, the error signal  $E_m(z)$  measured by the monitoring microphone is written as

$$E_{m}(z) = D_{m}(z) + S_{m}(z) Y(z)$$
  
= [P\_{m}(z) + S\_{m}(z) W\_{o}(z)] X(z), (4)

where  $D_m(z)$  is the disturbance signal at the monitoring microphone. The primary path  $P_m(z)$  and secondary path  $S_m(z)$  are the transfer functions from the noise source and control source to the monitoring microphone, respectively. The auxiliary filter H(z) is trained to estimate the error signal  $E_m(z)$  based on the reference signal X(z). Therefore,

$$H(z) = -\frac{E_m(z)}{X(z)} = -[P_m(z) + S_m(z)W_o(z)]B(X).$$
 (5)

In the control stage, the temporary microphone is removed from the target ZoQ. The control filter is retained to minimize the power of  $E_{m'}(z) + H(z) X'(z)$ , where  $E_{m'}(z)$  and X'(z) are the error and reference signals in the control stage, respectively. The yellow block indicates a perfect model of the secondary path that is assumed to be available to the controller in the control stage. The z domain expression of the converged control filter in the AF-VS method is thus written as

$$W_{AF}(z) = -\frac{P_{m'}(z) + H(z)}{S_{m'}(z)} B(X'), \qquad (6)$$

where the primary path  $P_{m'}(z)$  and secondary path  $S_{m'}(z)$  are the transfer functions from the noise source and control source to the monitoring microphone in the control stage, respectively.



Figure 3: Block diagram of the RM-VS method.

#### 2.2. Remote microphone based virtual sensing method

Figure 3 shows the block diagram of the RM-VS method in z domain. In the tuning stage, the disturbance signal measured by the monitoring microphone is used to estimate the disturbance signal measured by the virtual microphone. Such estimation is carried out by the relative primary path model  $C_p(z)$ , *i.e.* 

$$D_{v}(z) = C_{p}(z) D_{m}(z).$$
(7)

Considering the primary and virtual primary paths, we obtain

$$C_{p}(z) = \frac{P_{v}(z) X(z)}{P_{m}(z) X(z)} = \frac{P_{v}(z)}{P_{m}(z)} B(X).$$
(8)

When compared with the AF-VS method, the RM-VS method is likely to result in less noise reduction due to dips in the denominator.

In the control stage, the RM-VS method firstly estimates the disturbance signal at the monitoring microphone by

$$\hat{D}_{m'}(z) = E_{m'}(z) - S_{m'}(z) Y'(z), \qquad (9)$$

where Y'(z) is the control signal in the control stage. Secondly, the virtual error signal  $\hat{E}_{v'}(z)$  is estimated by

$$\hat{E}_{v'}(z) = C_p(z) \,\hat{D}_{m'}(z) + S_v(z) \,Y'(z) \,. \tag{10}$$

When the temporal microphones have been removed, the model of the virtual secondary path cannot be updated. And when a perfect model of the secondary path is assumed, the error signal measured by the monitoring microphone is expressed as

$$D_{m'}(z) = \hat{D}_{m'}(z) = P_{m'}(z) X'(z).$$
(11)

The control filter converges by minimizing the power of  $\hat{E}_v(z)$ . The z domain expression of the converged control filter of the RM-VS method is thus written as

$$W_{RM}(z) = -\frac{C_p(z) P_{m'}(z)}{S_v(z)} B(X').$$
(12)

When there is a change in the virtual secondary path, the accurate model of the virtual secondary path requested by the RM-VS method is not possible to be obtained with any online modeling technique, because no temporary microphone is placed at the target ZoQ during the control stage. In comparison, the AF-VS method requests the accurate model of the secondary path, which is available because online modeling technique can update the secondary path model based on the monitoring microphone.

#### 2.3. Relative path based virtual sensing method

To resolve the aforementioned disadvantage of the RM-VS method, we propose the RP-VS method. Figure 4 shows the additional tuning stage of the RP-VS method. As compared with the RM-VS method, the improvement of the RP-VS method is made by adding in another relative path modeling. The relative secondary path model  $C_s(z)$  is in charge of estimating the anti-noise signal at the virtual microphone based on the anti-noise signal measured by the monitoring microphone, *i.e.* 

$$S_{v}(z) Y(z) = C_{s}(z) S_{m}(z) Y(z).$$

$$(13)$$

After the tuning stage, we obtain

$$C_s(z) = \frac{S_v(z)}{S_m(z)} B(Y).$$
(14)



(b) Control stage

Figure 4: Block diagram of the RP-VS method.

In the control stage, the perfect model of the secondary path is assumed. Following a similar procedure of the RM-VS method, we estimate the virtual error signal  $\hat{E}_{v'}(z)$  as

$$\hat{E}_{v'}(z) = C_p(z)\,\hat{D}_{m'}(z) + C_s(z)\,S_{m'}(z)\,Y'(z)\,,\tag{15}$$

where the estimate of the disturbance signal at the monitoring microphone is still calculated by Eqs. (9) and (11). The z domain expression of the converged control filter of the RP-VS method is thus written as

$$W_{RP}(z) = -\frac{C_p(z) P_{m'}(z)}{C_s(z) S_{m'}(z)} B(X').$$
(16)

## 2.4. Comparison of virtual sensing methods

When there are no changes to the acoustic paths and noise characteristics, Eqs. (6), (12) and (16) all lead to the same optimal solution. However, fixing the coefficients of the control filter obtained in Eq. (2) would be the simplest way to control the noise in this case, rather than using any adaptive algorithms. Therefore, it is important to investigate the varying acoustic paths and varying noise characteristics for all the VS methods.

When the noise is assumed to occupy the entire bandwidth, *i.e.* letting B(X) = B(X') = B(Y) = 1, the control filters obtained in the control stage of the AF-VS method, the RM-VS method and the RP-VS method are manipulated as

$$W_{AF}(z) = \frac{P_m(z) - P_{m'}(z)}{S_{m'}(z)} + W_o \frac{S_m(z)}{S_{m'}(z)},$$
(17)

$$W_{RM}(z) = W_o \frac{P_{m'}(z)}{P_m(z)},$$
 (18)

Table 1: Z-transform analysis of different virtual sensing methods under conditions when there are changes to the primary and virtual primary paths, the secondary and virtual secondary paths, and all the acoustic paths. The notation (z) has been abbreviated.

$E_{v'}/X$	Invariant Secondary Paths Invariant Primary Pat		
	$(S_m = S_{m'} \text{ and } S_v = S_{v'})$	$(P_m = P_{m'} \text{ and } P_v = P_{v'})$	
FC Filter	$P_{v'} - P_v$	$\left(1 - \frac{S_{v'}}{S_v}\right) P_v$	
AF-VS Method	$P_{v'} - P_v + \frac{S_{v'}}{S_{m'}} \left( P_m - P_{m'} \right)$	$\left(1 - \frac{S_{v'}S_m}{S_vS_{m'}}\right)P_v$	
RM-VS Method	$P_{v'} - \frac{P_{m'}}{P_m} P_v$	$\left(1 - \frac{S_{v'}}{S_v}\right) P_v$	
RP-VS Method	$P_{v'} - \frac{P_{m'}}{P_m} P_v$	$\left(1 - \frac{S_{v'}S_m}{S_vS_{m'}}\right)P_v$	
$E_{v'}/X$	Varying Acoustic Paths		
FC Filter	$P_{v'} - \frac{S_{v'}}{S_v} P_v$		
AF-VS Method	$P_{v'} - \frac{S_{v'}S_m}{S_v S_{m'}} P_v + \frac{S_{v'}}{S_{m'}} (P_m - P_{m'})$		
RM-VS Method	$P_{v'} - \frac{S_{v'}P_{m'}}{S_v P_m} P_v$		
RP-VS Method	$P_{v'} - \frac{S_{v'} S_m P_{m'}}{S_v S_m / P_m} P_v$		

and

$$W_{RP}(z) = W_o \frac{S_m(z) P_{m'}(z)}{P_m(z) S_{m'}(z)},$$
(19)

respectively. We notice that when there is no change in the primary path, *i.e.*  $P_{m'}(z) = P_m(z)$ , the AF-VS method and the RP-VS method lead to the same solution. When there is no change in the secondary path, *i.e.*  $S_{m'}(z) = S_m(z)$ , the RM-VS method and the RP-VS method lead to the same solution.

Table 1 shows the ratio between the sound pressure level after control and the reference signal in the z domain. For the simplicity of presentations, the notation (z) has been abbreviated. When the secondary and virtual secondary paths are invariant, the FC filter remains effective if the change in the primary path is slight. Similarly, the AF-VS method remains effective if changes in both the primary and virtual primary paths are slight. In comparison, the performance of the RM-VS method and the RP-VS method relies only on the relative change of the primary and virtual primary paths. The optimal noise reduction is achieved when  $\frac{P_{v'}}{P_v} = \frac{P_{m'}}{P_m}$ . This condition is likely to be fulfilled in practice when the change of acoustics paths is incurred by the noise source itself. When the primary and virtual primary paths are invariant, the FC filter and the RM-VS method have the same performance, which depends on the change in the virtual secondary path. Meanwhile, the AF-VS method and the RP-VS method have the same performance, which is proportional to the relative change in the secondary and virtual secondary paths. The optimal noise reduction is achieved when  $\frac{S_{w'}}{S_v} = \frac{S_{m'}}{S_m}$ . This condition may happen when the change of acoustic paths is due to the control source. When all the acoustic paths are varying, the RP-VS method still achieves the optimal noise reduction, so long as the relative changes in the acoustic paths are balanced.

Similarly, the noise level after control can be analyzed for a MCFFANC system, which for instance consists of I reference microphones, J secondary loudspeakers, K monitoring microphones and L virtual microphones. For the simplicity of presentations, the notation (z) has been abbreviated and the noise is assumed to occupy the entire bandwidth. The dimensions of matrixes and vectors are explicitly marked on the superscript.

In the training stage, the optimum control filter  $\mathbf{W}_o$  is written as

$$\mathbf{W}_{o}^{(J \times I)} = -\left[\mathbf{S}_{v}^{(L \times J)}\right]^{\dagger} \mathbf{P}_{v}^{(L \times I)},\tag{20}$$

where  $\mathbf{P}_{v}$  is the virtual primary path;  $\mathbf{S}_{v}$  is the virtual secondary path; and † denotes the pseudo inverse of a matrix. The AF-VS method obtains the auxiliary filter  $\mathbf{H}$  as

$$\mathbf{H}^{(K \times I)} = -\mathbf{P}_m^{(K \times I)} + \mathbf{S}_m^{(K \times J)} \mathbf{W}_o^{(J \times I)},$$
(21)

where  $\mathbf{P}_m$  and  $\mathbf{S}_m$  are the primary and secondary paths, respectively. The RM-VS method obtains the relative primary path model  $\mathbf{C}_p$  as

$$\mathbf{C}_{p}^{(L\times K)} = \mathbf{P}_{v}^{(L\times I)} \left[\mathbf{P}_{m}^{(K\times I)}\right]^{\dagger}, \qquad (22)$$

while the RP-VS method in addition obtains the relative secondary path model  $\mathbf{C}_s$  as

$$\mathbf{C}_{s}^{(L\times K)} = \mathbf{S}_{v}^{(L\times J)} \left[\mathbf{S}_{m}^{(K\times J)}\right]^{\dagger}.$$
(23)

In the control stage, the control filters of the AF-VS method, the RM-VS method and the RP-VS method are respectively written as

$$\mathbf{W}_{AF}(z) = \left[\mathbf{S}_{m'}^{(K\times J)}\right]^{\dagger} \left[\mathbf{P}_{m}^{(K\times I)} - \mathbf{P}_{m'}^{(K\times I)} + \mathbf{S}_{m}^{(K\times J)}\mathbf{W}_{o}^{(J\times I)}\right], \quad (24)$$

$$\mathbf{W}_{RM}\left(z\right) = \mathbf{W}_{o}^{\left(J \times I\right)} \left[\mathbf{P}_{m}^{\left(K \times I\right)}\right]^{\dagger} \mathbf{P}_{m'}^{\left(K \times I\right)},\tag{25}$$

and

$$\mathbf{W}_{RP}\left(z\right) = \left[\mathbf{S}_{m'}^{(K\times J)}\right]^{\dagger} \mathbf{S}_{m}^{(K\times J)} \mathbf{W}_{o}^{(J\times I)} \left[\mathbf{P}_{m}^{(K\times I)}\right]^{\dagger} \mathbf{P}_{m'}^{(K\times I)},$$
(26)

where  $\mathbf{P}_{m'}$  and  $\mathbf{S}_{m'}$  are the primary and secondary paths that might be different from those in the training stage. Furthermore, the virtual primary and secondary paths in the control stage are denoted as  $\mathbf{P}_{v'}$  and  $\mathbf{S}_{v'}$ , respectively.

Table 2 shows the noise levels after control  $\mathbf{E}_{v'}$  of the AF-VS method, the RM-VS method, the RP-VS method and the FC filter for comparison, where  $\mathbf{X}$  is the reference signal vector with the size of  $I \times 1$ . Observations are generally similar to the single-channel ANC system. However, it is worth noting

Table 2: Z-transform analysis of different virtual sensing methods under conditions when the primary path change, the secondary path change, and all the path change in a MCF-FANC system. I denotes the identity matrix.

$\ \mathbf{E}_{v'}\ $	Invariant Secondary Paths	Invariant Primary Paths		
	$(\mathbf{S}_m = \mathbf{S}_{m'} \text{ and } \mathbf{S}_v = \mathbf{S}_{v'})$	$(\mathbf{P}_m = \mathbf{P}_{m'} \text{ and } \mathbf{P}_v = \mathbf{P}_{v'})$		
FC Filter	$\ (\mathbf{P}_{v'}-\mathbf{P}_v)\mathbf{X}\ $	$\left\  \left[ \mathbf{I} - \mathbf{S}_{v'} \left( \mathbf{S}_{v}  ight)^{\dagger}  ight] \mathbf{P}_{v} \mathbf{X}  ight\ $		
AF-VS Method	$\left\  \left[ \mathbf{P}_{v'} - \mathbf{P}_{v} + \mathbf{S}_{v'} \left( \mathbf{S}_{m'} \right)^{\dagger} \left( \mathbf{P}_{m} - \mathbf{P}_{m'} \right) \right] \mathbf{X} \right\ $	$\left\  \left[ \mathbf{I} - \mathbf{S}_{v'} \left( \mathbf{S}_{m} \right)^{\dagger} \mathbf{S}_{m'} \left( \mathbf{S}_{v} \right)^{\dagger} \right] \mathbf{P}_{v} \mathbf{X} \right\ $		
RM-VS Method	$\begin{bmatrix} \mathbf{P}_{v'} - \mathbf{P}_{v} \left( \mathbf{P}_{m} \right)^{\dagger} \mathbf{P}_{m'} \end{bmatrix} \mathbf{X}$	$\left[ \mathbf{I} - \mathbf{S}_{v'} \left( \mathbf{S}_{v} \right)^{\dagger} \right] \mathbf{P}_{v} \mathbf{X}$		
RP-VS Method	$\left\  \left[ \mathbf{P}_{v'} - \mathbf{P}_{v} \left( \mathbf{P}_{m} \right)^{\dagger} \mathbf{P}_{m'} \right] \mathbf{X} \right\ $	$\left\  \left[ \mathbf{I} - \mathbf{S}_{v'} \left( \mathbf{S}_{m} \right)^{\dagger} \mathbf{S}_{m'} \left( \mathbf{S}_{v} \right)^{\dagger} \right] \mathbf{P}_{v} \mathbf{X} \right\ $		
$\ \mathbf{E}_{v'}\ $	Varying Acousti	c Paths		
FC Filter	$\left\  \begin{bmatrix} \mathbf{P}_{v'} - \mathbf{S}_{v'} \left( \mathbf{S}_{v} \right)^{\dagger} \mathbf{P}_{v} \end{bmatrix} \mathbf{X} \right\ $			
AF-VS Method	$\left\  \begin{bmatrix} \mathbf{P}_{v'} - \mathbf{S}_{v'} (\mathbf{S}_m)^{\dagger}  \mathbf{S}_{m'} (\mathbf{S}_v)^{\dagger}  \mathbf{P}_v + \mathbf{S}_{v'} \begin{bmatrix} \mathbf{S}_{v'} & \mathbf{S}_{v'} \end{bmatrix} \right\ $	$\mathbf{S}_{v'} \left( \mathbf{S}_{m'} \right)^{\dagger} \left( \mathbf{P}_m - \mathbf{P}_{m'} \right) \mathbf{X}$		
RM-VS Method	$\left\  \begin{bmatrix} \mathbf{P}_{v'} - \mathbf{S}_{v'} \left( \mathbf{S}_{v} \right)^{\dagger} \mathbf{P}_{v} \left( \mathbf{P}_{m} \right)^{\dagger} \mathbf{P}_{m'} \end{bmatrix} \mathbf{X} \right\ $			
RP-VS Method	$\begin{bmatrix} \mathbf{P}_{v'} - \mathbf{S}_{v'} (\mathbf{S}_m)^{\dagger}  \mathbf{S}_{m'} (\mathbf{S}_v)^{\dagger} \end{bmatrix}$	$\mathbf{P}_{v} \left( \mathbf{P}_{m} \right)^{\dagger} \mathbf{P}_{m'} \mathbf{X}$		

that the effectiveness of all VS methods cannot be guaranteed under arbitrary path changes. In the multi-channel ANC system, the balanced relative path change happens much more occasionally than that in the single-channel ANC system, even for just a fraction of the noise bandwidth. Therefore, multi-channel ANC systems are big challenges for all the VS methods.

# 3. Simulation Results

In this section, we establish two experimental setups of the single-channel and dual-channel feedforward ANC systems. The acoustic paths are acquired by a real-time digital signal processor (DSP) platform. Three changes in different acoustic paths are measured. For the primary and virtual primary paths, the impulse responses are measured up to 75 ms. The length of the secondary and virtual secondary paths is 25 ms. The sampling rate is 16



Figure 5: Case (1,1,1) single-channel ANC system setup.

kHz. The memory length of the control filter, auxiliary filter, and relative path models is 400 taps. In different VS methods, all the adaptive algorithms adopt the normalized step size of 0.01. Noise reduction levels are calculated after control filters completely converge.

# 3.1. Single-channel feedforward ANC system

Figure 5 shows the setup of a case (1,1,1) single-channel feedforward ANC system and variations of acoustic paths. The frequency band of the noise source is from 400 Hz to 1600 Hz. Changes in the acoustic paths are made by manually moving the microphones 12 cm away from their original positions.



Figure 6: Magnitude and phase responses of acoustic paths used in the single-channel ANC simulation.

Noise Reduction Level	FC Filter	AF-VS Method	RM-VS Method	RP-VS Method
Tuning Condition	24.3	24.2	23.7	23.7
Primary and Virtual Primary Paths I	4.2	6.4	14.0	14.0
Primary and Virtual Primary Paths II	15.1	14.7	13.3	13.3
Primary and Virtual Primary Paths III	15.8	16.5	15.3	15.3
Secondary and Virtual Secondary Paths I	8.0	12.3	7.9	12.2
Secondary and Virtual Secondary Paths II	11.8	12.5	11.8	12.5
Secondary and Virtual Secondary Paths III	7.4	13.8	7.3	13.9
All of the Acoustic Paths I	14.7	10.3	8.4	14.5
All of the Acoustic Paths II	10.9	10.9	9.0	10.5
All of the Acoustic Paths III	7.6	12.7	7.0	11.8

Table 3: Noise reduction levels at the target ZoQ of the single-channel feedforward ANC system with varying acoustic paths.

As the monitoring microphone and the virtual microphone are fastened to separate microphone stands, they are not moved into the same direction. Three groups of measurements are carried out to record all the acoustic paths. They are then used as the primary and virtual primary path changes, secondary and virtual secondary path changes, and all path changes in the simulation. Figure 6 further exhibits the magnitude and phase responses of these acoustic paths. There is an obvious dip in primary paths at about 825 Hz.

Table 3 lists noise reduction levels of different VS methods at the target ZoQ. When there are no changes in all of the acoustic paths, it is labeled as the tuning condition. The AF-VS method achieves nearly the same performance as the FC filter, which is the optimal solution in this condition. Both the RM-VS method and the RP-VS method result in less noise reduction due to the dip in the frequency response of the primary path. When

there are changes in the primary and virtual primary paths, the RM-VS method and the RP-VS method achieve the same levels of noise reduction. Although their performance is not optimal, the RM-VS method and the RP-VS method show significant robustness as compared to the AF-VS method, and far better performance than the FC filter. When there are changes in the secondary and virtual secondary paths, the AF-VS method and the RP-VS method have very similar noise reduction levels. They outperform the FC filter and the RM-VS method. The latter two methods also have very similar noise reduction levels. These observations are consistent with the theoretical analysis in Table 1. Furthermore, when all of the acoustic paths are varying, the RM-VS method obtains the least noise reduction and the RP-VS method outperforms the other VS methods in terms of the average noise reduction level. None of the VS methods are guaranteed to provide the perfect performance in all circumstances.

When there are no changes in all of the acoustic paths, the noise frequency band is firstly changed to a tuning band (from 600 Hz to 1200 Hz), and then shifted to five different testing frequency bands. Noise reduction levels of the FC filter and VS methods at the target ZoQ are shown in Fig. 7. All of the aforementioned methods obtain less noise reduction levels when the noise frequency band varies. The FC filter even causes increments in the sound pressure level, when the testing frequency band is outside the tuning frequency band. The control filters obtained in the control stage of the RP-VS method have the closest phase responses to those of the optimal control filters. This is likely due to the fact that the RP-VS method has trained the relative primary path model  $C_p(z)$  and the relative secondary



Figure 7: Noise reduction levels at the target ZoQ of the single-channel ANC system with varying noise frequency bands.

path model  $C_s(z)$  separately. Since phase plays a more significant role in ANC rather than magnitude, the RP-VS method shows the most robust performance when the noise frequency band varies in the simulation. The change in the noise frequency band is of particular importance to the RM-VS method and the RP-VS method. This is because the noise source is uncontrollable. The relative primary path model is favorable to be trained under one working condition of the noise source, but there may be many other working conditions.

# 3.2. Dual-channel feedforward ANC system

Figure 8 shows the setup of a case (1,2,2) dual-channel feedforward ANC system and variations of cross-channel acoustic paths. The frequency band of the noise source is from 400 Hz to 1600 Hz. Changes in the acoustic paths are made by manually moving microphones 12 cm away from their original

positions. Moreover, Fig. 9 exhibits the magnitude and phase responses of those cross-channel acoustic paths.

Table 4 lists noise reduction levels of different VS methods at the target ZoQ of the dual-channel ANC system. For the tuning condition, the AF-VS method still outperforms the RM-VS method and the RP-VS method. However, the difference in their performance is reduced as compared to the single-channel ANC system. When there are changes in the primary and virtual primary paths, the RM-VS method and the RP-VS method exhibit more robustness than the FC filter and the AF-VS method. For the third combinations of the primary and virtual primary paths, although the AF-VS method achieves the highest noise reduction level, the other three methods can also achieve noise reduction levels of about 15 dB. When there are changes in the secondary and virtual secondary paths, the AF-VS method and the RP-VS method no longer result in the same noise reduction levels. The cross-channel acoustic paths incur errors in the relative secondary path models that are used by the RP-VS method. However, with more computational power consumed and more prior information obtained, the RP-VS method still achieves the best average noise reduction performance in the simulation.

Similar to the simulation results of the single-channel ANC system, all the methods obtain less noise reduction in the dual-channel ANC system when the noise frequency band varies. The tuning frequency band is from 600 Hz to 1200 Hz. Two of the five testing frequency bands are totally outside the tuning frequency band. One testing frequency band is broader than the tuning frequency band. The other two testing frequency band overlap with



Figure 8: Case (1,2,2) dual-channel ANC system setup.



Figure 9: Magnitude and phase responses of cross-channel acoustic paths used in the dual-channel ANC simulation.

Noise Reduction Level	FC Filter	AF-VS Method	RM-VS Method	RP-VS Method
Tuning Condition	23.7	23.7	23.0	23.1
Primary and Virtual Primary Paths I	10.0	11.8	13.7	13.7
Primary and Virtual Primary Paths II	8.3	9.8	14.3	14.3
Primary and Virtual Primary Paths III	15.4	16.1	14.4	14.4
Secondary and Virtual Secondary Paths I	4.2	5.4	4.2	6.2
Secondary and Virtual Secondary Paths II	7.7	10.4	7.6	9.9
Secondary and Virtual Secondary Paths III	5.7	5.0	5.6	5.8
All of the Acoustic Paths I	4.9	6.8	4.3	6.0
All of the Acoustic Paths II	10.0	8.8	7.3	7.9
All of the Acoustic Paths III	5.3	4.7	4.7	5.5

Table 4: Noise reduction levels at the target ZoQ of the dual-channel ANC system with varying acoustic paths.

the tuning frequency band. Noise reduction levels of the FC filter and VS methods at the target ZoQ are shown in Fig. 10. The FC filter and the AF-VS method cause increments in the sound pressure level when the testing frequency band is lower than the tuning frequency band. Among all the VS methods, the RP-VS method leads to the control filters with the closest phase responses to the optimal control filters in the control stage. This gains the advantage of the RP-VS method when dealing with the varying noise frequency band in the simulation. However, the noise reduction levels achieved by the RP-VS method in the dual-channel ANC system are less than those in the single-channel ANC system. Similar trends are observed for the RM-VS method too. The difficulty caused by the cross-channel acoustic paths is hence demonstrated.



Figure 10: Noise reduction levels at the target ZoQ of the dual-channel ANC system with varying noise frequency bands.

# 4. Experiment results of an ANC casing

The ANC casing is built up with a case (1,4,4) feedforward ANC control system. It consists of one reference microphone inside the casing, four loudspeakers as the control sources, four monitoring microphones fastened near the loudspeakers and four removable virtual error microphones that are supported by racks in the tuning stage. The 3D design model and prototype photo of the ANC casing are shown in Fig. 11. A computer fan, whose full speed is about 2500 revolutions per minute, is enclosed in the ANC casing for the experiment. The ANC controller is a real-time DSP platform supporting four channels of digital-to-analog conversion and eight channels of analogto-digital conversion. Therefore, in the tuning stage of the RM-VS method and the RP-VS method, only the monitoring microphones and the virtual microphones are connected to the DSP platform. The relative primary path models are estimated with the noise emitted from the computer fan. The relative secondary path models are obtained by turning off the computer fan and playing white noise from each secondary loudspeaker. Subsequently, the reference microphone and the virtual microphones are connected to the DSP platform, in order for the optimal control filters to be obtained by the FxLMS algorithm. With the control filter coefficients fixed, the monitoring microphones are connected to replace the virtual microphones and to tune the auxiliary filters of the AF-VS method. In the control stage, the virtual microphones are connected to another DSP platform to record the sound pressure level at the target ZoQ. The AF-VS method, the RM-VS method and the RP-VS method all adopt the reference microphone and the monitoring microphones to update their control filter coefficients. The speed of the computer fan is adjusted during the control stage. For each speed setting, the FxLMS baseline is obtained by using the reference microphone and the virtual microphones. The circuit configuration of the ANC casing in the experiment is shown in Fig. 12.

In the tuning stage, the computer fan is set to 30% of its full speed. The sound pressure level at the target ZoQ is 58 dB, while the floor noise of the measurement room is 34 dB. The relative path models are trained with the fan noise on the real-time DSP platform. The FxLMS baseline is then trained with the reference microphone and four virtual microphones. In the control stage, the virtual microphones are only used to showcase the noise reduction performance. The noise power spectra are plotted in Fig. 13. It shows that the ANC casing achieves broadband noise reduction at the target ZoQ. The performance of the AF-VS method is very close to the FxLMS baseline,



Figure 11: 3D design model (left) and prototype photo (right) of the ANC Casing.

while the RM-VS method and the RP-VS method provide competitive noise reduction performance. Next, the fan speed is accelerated to its full speed. The noise frequency band is not significantly changed. However, the sound pressure level at the target ZoQ is increased to 73 dB. The computer fan causes vibration of the ANC casing, which leads to fluctuations in acoustic paths. Without further tuning of the auxiliary filters and the relative path models, the RP-VS method is as effective as the AF-VS method and the RM-VS method. They all achieve notable broadband noise reduction, but none of the VS methods can outperform the other two methods substantially.

Figure 15 shows the convergence curves of the VS methods and the FxLMS baseline at the target ZoQ and the monitoring microphones when the fan speed is 30%, 60%, 80% and 100% of its full speed. The relative path



Figure 12: Circuit configuration of the ANC casing in the experiment.



Figure 13: Noise spectra at the virtual microphone and the monitoring microphone when the fan speed is at 30% of its full speed. Note that this is also the tuning condition for the AF-VS method, the RM-VS method and the RP-VS method.



Figure 14: Noise spectra at the virtual microphone and the monitoring microphone when the fan speed is at 100% of its full speed.

models are only obtained with the tuning condition when the fan speed is at 30% of its full speed. It is worth noting that with higher fan speed, more noise reduction is observed. This is due to the floor noise of the measurement room that hinders the convergence of the adaptive algorithm. Furthermore, the sound pressure level at the monitoring microphone is increased when the target ZoQ is formed. This demonstrates the necessity of VS methods in applications with the ANC casing. The monitoring microphones are placed very close to the control sources for better appearance and safer use. However, the target ZoQ can be far away from the control sources. VS methods are effective solutions to this practical situation. Besides the AF-VS method and the RM-VS method, the RP-VS method is a new option that is suitable when dealing with varying acoustic paths and varying noise characteristics.



Figure 15: Convergence curves at the virtual microphone and the monitoring microphone with different fan speed settings.

# 5. Conclusions

The two most commonly applied VS methods, namely the AF-VS method and the RM-VS method, are compared with the newly proposed RP-VS method in this paper through analytical analysis and simulations. An investigation of varying acoustic paths and noise characteristics is highlighted. Simulation results, obtained with case (1,1,1) and case (1,2,2) feedforward ANC systems, are in good agreement with the analytical results.

The RP-VS method can behave in the same way as the AF-VS method or the RM-VS method, under specific assumptions of varying acoustic paths. When the secondary and virtual secondary paths are invariant, the RM-VS method and the RP-VS method are likely to provide more noise reduction than the AF-VS method. When the primary and virtual primary paths are invariant, the AF-VS method and the RP-VS method can make use of the online secondary path modeling technique to achieve better noise reduction performance. Moreover, all the VS methods demonstrate their robustness, which is a lack in the FC filter due to its non-adaptivity. With more crosschannel acoustic paths involved, multi-channel ANC systems are more sensitive to varying acoustic paths, which consequently are more challenging in practical implementations.

Lastly, an ANC casing is built up with a case (1,4,4) feedforward ANC system, implementing the RP-VS method on a real-time DSP platform. The tuning stage is carried out with a relatively low fan speed, while the control stage deals with higher speeds and correspondingly higher noise levels. The experiment results validate that the RP-VS method is as effective as the AF-VS method and the RM-VS method in the implementation of the ANC casing. Therefore, the RP-VS method could be an alternative VS method to deal with varying acoustic paths and varying noise characteristics.

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