MMSE Filter Design for Full-duplex Filter-and-forward MIMO Relays under Limited Dynamic Range

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Abstract

We study the problem of optimizing the end-to-end performance of a full-duplex filter-and-forward MIMO relay link, consisting of a source, a relay, and a destination node, by employing linear filtering at each node. The system model accounts for multipath propagation and self-interference at the relay, as well as transmitter impairments and limited dynamic range at every node. The design accommodates signals with arbitrary spectra and includes the direct link between the source and destination nodes. Under the minimum mean square error criterion, the resulting non-convex problem is approximated by a sequence of convex problems and solved by means of an alternating minimization method. Linear constraints allocate some of the degrees of freedom in the relay to guarantee a sufficiently small residual self-interference. Simulations quantify the impact of degrees of freedom, the dynamic range, and the balance between direct and relay paths on the link performance.

I. INTRODUCTION

Relays expand the source-destination architecture by extending coverage area and improving end-to-end performance [1], [2]. Among the different relay protocols, filter-and-forward (FF) relaying constitutes an attractive alternative, in terms of complexity and performance, to other relaying techniques such as amplify-and-forward (AF), decode-and-forward (DF) or compress-and-forward (CF) [3]–[7]. In contrast to AF relaying, where the signal is forwarded after a spatial transformation, FF relaying forwards the signal after passing it through a linear filter, usually a linear finite impulse response (FIR) filter [8]–[12]. The spectrum shaping capabilities of FF relaying bring about a performance edge over its AF counterpart, as reported in [13]. Additionally, FF relaying allows for a scalable design in comparison to the implementation of DF and CF relaying (normally comprising, among others, timing recovery, frame alignment and/or signal regeneration operations), because the number of parameters grows linearly with the number of antennas and the number of filter taps. By avoiding the decoding and re-encoding of the signal, AF and FF relays are able to comply with the strict latency requirements of 5G systems.

In combination with a full-duplex (FD) protocol, any efficient relay design must deal with the presence of self-interference (SI) distortion which is a consequence of simultaneous transmission and reception in the same frequency [14]–[20]. If not properly mitigated, SI severely impacts the performance, as it has been reported in [3], [21]. To cope with SI distortion, different mitigation methods, both in the analog and digital domains, have been developed, from which we highlight those based on interference suppression (exploitation of spatial diversity), e.g., [14], [22]–[25], and those based on interference cancellation (generation of an SI replica), e.g., [26]–[30]. Due to insufficient mitigation and high power transmission of the relay, residual SI may still be strong enough to limit performance [14], and, therefore, the relay design must account for the presence of residual SI.

Any relay implementing a linear filtering based FF protocol, under the presence of residual SI, is subjected to an impulse response of infinite duration, in which the relay operation is contained inside the feedback loop caused by the SI [29]. As a consequence, data signal and SI are correlated, which turns any optimal design strategy into an intractable problem and possibly leading to an unstable system. This is in contrast to DF and CF relays, where the processing delay, defined as the relay input-to-output time delay, is long enough to decorrelate the data signal and the SI [30].

Decorrelation between data and SI can be achieved by introducing additional processing delay, i.e., the relay waits a sufficiently long number of samples before transmission. This decorrelation property applies in relays operating in frequency domain, where the processing delay is at least one symbol or several time samples [14]. Under those conditions, the relay design results in a tractable problem. Processing delay is an important parameter, and in the case of an orthogonal frequency-division multiplexing (OFDM) system it should not exceed the cyclic prefix duration to avoid inter-symbol interference due to the direct link. We propose a time-domain design in which no additional delay is introduced in the processing path. Still, the method ensures decorrelation between data signal and SI by transmitting in the nullspace of the SI channel.

SI distortion is not the only performance-limiting factor in an FD architecture. Another important cause of performance degradation is the limited dynamic range at reception and transmission [31]–[33]. Limited dynamic range is due to imperfections in analog/digital conversion and nonlinear effects in the modulation/demodulation process [31], and translates into additional
distortion in the system. Our design method incorporates limited dynamic range at every node of the link as additional parameters of the system model.

A. Contributions of the paper

We consider a MIMO FD FF relay link with limited dynamic range and residual SI distortion after antenna isolation/analog cancellation at the relay [17]. We introduce linear filters at each node, which are designed by minimizing the mean square error (MSE) at the destination node. The resulting optimization problem is non-convex and the IIR nature of the relay makes the problem intractable. We transform the original problem into a tractable one by imposing interference-free constraints on the relay. This modified problem is solved by employing an alternating optimization technique. In particular, node filters are obtained by solving a sequence of decoupled convex optimization problems.

Existing literature in FF design [13], [34]–[36] only considers half-duplex protocol and excludes direct link between source and destination [34], whereas we model a FD system with direct link between source and destination. Although [37] contemplates the presence of a direct link between source and destination, it does not account for a FD relay with SI or limited dynamic range at each node of the link. Our design incorporates additional noise sources that model limited dynamic range at both transmission and reception.

In [38], the authors study the problem of an FF FD relay network. In contrast to our approach, the system model considers filters in the frequency domain, where there is no correlation between data signal and SI, and dynamic range is infinite. Furthermore, frequency-domain processing introduces an unavoidable of at least one OFDM symbol. While [39], [40] consider source-destination link, only the case of single antenna with flat-frequency channel response and infinite dynamic range is treated.

In [41], the authors consider an FF FD relay network in the frequency domain with limited dynamic range and direct link between source and destination. A gradient-projection based approach is used to maximize the signal-to-noise-plus-interference ratio. Our design method works in the time-domain and is, therefore, modulation independent, and able to deal with multipath distortion as well as reducing the end-to-end delay because it does not require synchronization at the relay.

Finally, [31] considers the problem of limited dynamic range for a DF relay link assuming uncorrelated input and output signals. Concretely, we consider the correlation between data signal and self-interference at the FF relay, which has a major impact on the filter design, particularly in the assignment of its degrees of freedom.

B. Organization of the paper

The paper is organized as follows. Section II describes the system model of the relay link, with special attention to the limited dynamic range of the nodes and the equivalent impulse response of the link. Section III formulates the filter design problem based on the minimum MSE (MMSE) criterion and details the required approximations in order to obtain a tractable optimization problem. Section IV solves the optimization problem by means of an alternating optimization approach, in which every filter is the solution to a convex optimization problem. Section V illustrates the performance of the design algorithm for an OFDM relay system. Finally Section VI draws the conclusions.

C. Notation

Let \( \{A[k]\}_{k=0}^{L_A} \), with \( A[k] \in \mathbb{C}^{M \times N} \), denote the impulse response of a complex-valued, \( L_A \)-th-order causal FIR MIMO filter. The row-expanded matrix of \( A[n] \), denoted by \( \mathbf{A} \), is defined as \( \mathbf{A} = [A[0] \ldots A[L_A]] \in \mathbb{C}^{M \times N(L_A+1)} \). We define the squared Frobenius filter norm as \( \|A\|^2 = \text{tr}(A^H A) = \sum_{k=0}^{L_A} \text{tr}(A[k]A^H[k]) \).

The convolution between filters \( A[n] \) and \( B[n] \), of respective sizes \( M \times N \) and \( N \times P \) and orders \( L_A \) and \( L_B \), yields filter \( C[n] = A[n] \ast B[n] = \sum_{k=0}^{L_C} A[k]B[n-k] \), of size \( M \times P \) and order \( L_C = L_A + L_B \). In its row-expanded matrix, \( C[n] \) is expressed as

\[
C = A \mathcal{R}(B)
\]

(1)

where \( \mathcal{R}(B) \in \mathbb{C}^{N(L_A+1) \times P(L_C+1)} \) is the row-diagonal-expanded matrix of order \( L_A \) of \( B[n] \) [42, Sec.7.5.],

\[
\mathcal{R}(B) = \begin{bmatrix}
B[0] & \ldots & B[L_B] & 0_{N \times P} & \ldots & 0_{N \times P} \\
0_{N \times P} & B[0] & \ldots & B[L_B] & \ldots & 0_{N \times P} \\
\vdots & \ddots & \ddots & \ddots & \ddots & \vdots \\
0_{N \times P} & \ldots & 0_{N \times P} & B[0] & \ldots & B[L_B]
\end{bmatrix}
\]

(2)

where \( 0_{N \times P} \) denotes a null matrix of size \( N \times P \). Note that \( \mathcal{R}(C) = \mathcal{R}(A) \mathcal{R}(B) \) for a proper size of the expanded matrices. The column-diagonal-expanded matrix of order \( L_A \) of \( B[n] \), denoted by \( \mathcal{C}(B) \in \mathbb{C}^{N(L_C+1) \times P(L_A+1)} \), is defined as [42,
The system model of an FF relay link incorporating the proposed design is depicted in Fig. 1 and consists of a source node (S) equipped with M_S antennas that transmits the signal \( s[n] \), a destination node (D) equipped with N_D antennas that receives the signal \( d[n] \), and a relay node (R) equipped with N_R antennas that receives and transmits the signal \( r[n] \). Assuming a block fading model, the \( L_{ij} \)-th order channel between node \( i \) and node \( j \), where \( i \in \{S, R\} \) and \( j \in \{R, D\} \), is denoted by \( H_{ij}[n] \in \mathbb{C}^{N_{ij} \times M_i} \). Note that the strictly causal \( H_{RR}[n] \) with coefficients \( \{H_{RR}[k]\}_{k=1}^{L_{RR}} \) is the residual SI channel after antenna isolation/analogue cancellation [6, 17, 43]. Mitigation in the analog domain eliminates a significant part of the self-interference. The mitigation level is a function of the frequency and normally insufficient for a wideband signal. Impulse responses \( H_{ij}[n] \) account for the analog front-end distortion and the propagation effects between nodes, as well as their respective delays. Noise sources are represented by vectors \( v_{i}[n] \in \mathbb{C}^{M_i} \), \( w_j[n] \in \mathbb{C}^{N_j} \), and \( n_{i}[n] \in \mathbb{C}^{N_i} \), while aggregated noise sources containing all noise terms at a node are represented by vectors \( z_j[n] \in \mathbb{C}^{N_j} \), where \( i \in \{S, R\} \) and \( j \in \{R, D\} \). As seen from Fig. 1, each node filters the locally observed signal. The design criterion for \( g_S[n] \), \( g_D[n] \), and \( G_R[n] \) is to minimize the MMSE given by \( \mathbb{E}\{|d[n] - s[n - \tau]|^2\} \).
with \( \tau \geq 0 \) being a design parameter. The joint design of these filters under the MSE criterion is the topic of subsequent sections of this paper.

Starting with the signal in S, from Fig. 1, the \( L_S \)th-order filter \( g_S[n] \in \mathbb{C}^{M_S} \) precodes the data signal \( s[n] \)
\[
\hat{s}[n] = g_S[n] * s[n]
\] (6)
The relay implements an FF protocol, modeled as the \( L_R \)th-order filter \( G_R[n] \in \mathbb{C}^{M_R \times N_k} \) and transmits the signal
\[
\hat{r}[n] = G_R[n] * r[n]
\] (7)
where the input signal to the relay filter \( r[n] \) can be decomposed into
\[
r[n] = \hat{r}[n] + \underbrace{H_{RR}[n] * \hat{r}[n]}_{\text{self-interference i}[n]} + \underbrace{n_R[n] + w_R[n] + H_{RR}[n] * v_R[n]}_{\text{noise terms}}
\] (8)
with \( \hat{r}[n] \in \mathbb{C}^{N_k} \) being the incoming source signal, i.e.,
\[
\hat{r}[n] = H_{SR}[n] * (\hat{s}[n] + v_S[n])
\] (9)
and \( i[n] \) denoting the self-interference. The presence of the self-interference path introduces a feedback loop, due to which the relay becomes an infinite impulse response (IIR) linear MIMO filter. The impulse response of the relay \( G_{eq}[n] \in \mathbb{C}^{M_k \times N_k} \) is given by
\[
G_{eq}[n] = \langle I_{M_R} \delta[n] - G_R[n] * H_{RR}[n] \rangle^{-1} * G_R[n] = G_R[n] * \langle I_{N_k} \delta[n] - H_{RR}[n] * G_R[n] \rangle^{-1}
\] (10)
The \( L_D \)th-order filter \( g_D[n] \in \mathbb{C}^{N_D} \) linearly combines the received signal \( d[n] \) at D, and produces the output
\[
d[n] = g_D[n] * d[n]
\] (11)
from which source signal \( s[n] \) is recovered. We may now rewrite (11), in terms of the various channels and noise sources, as follows
\[
d[n] = g_D[n] * (H_{eq}[n] * g_S[n] * s[n] + z_D[n])
\] (12) with
\[
H_{eq}[n] = H_{SD}[n] + H_{RD}[n] * G_{eq}[n] * H_{SR}[n]
\] (13)
\[
\] (14)
The \( L_{eq} \)th-order filter \( H_{eq}[n] \in \mathbb{C}^{N_D \times M_k} \) represents the overall channel from S to D including direct (S-D) and relay (S-R-D) paths. From (12)-(14) we see that synchronization between direct-link and source-relay-destination signals is not needed as long as the delay spread of \( H_{eq}[n] \) is smaller than the cyclic prefix length.

From (10), \( L_{eq} \to \infty \), as the equivalent channel response has an infinite duration in principle. Vector \( z_D[n] \in \mathbb{C}^{N_D} \) in (12) comprises all noise terms at D and \( z_R[n] \in \mathbb{C}^{M_k} \) denotes the aggregated noise sources at the relay output:
\[
z_R[n] = G_{eq}[n] * (H_{SR}[n] * v_S[n] + n_R[n] + w_R[n]) + \langle I_{M_k} \delta[n] - G_R[n] * H_{RR}[n] \rangle^{-1} * v_R[n]
\] (15)
where vectors \( n_R[n] \in \mathbb{C}^{N_k} \) and \( n_D[n] \in \mathbb{C}^{N_D} \) are the thermal noise sources in R and D, respectively. Their respective distributions are
\[
n_R[n] \sim \mathcal{CN}(0, \sigma_R^2 I)
\] (16)
\[
n_D[n] \sim \mathcal{CN}(0, \sigma_D^2 I)
\] (17)
Vectors \( v_S[n] \in \mathbb{C}^{M_S} \) and \( v_R[n] \in \mathbb{C}^{M_k} \) model imperfections at the transmit sides of S and R, and are assumed normally distributed and statistically independent of the transmit signals \( \hat{s}[n] \) and \( \hat{r}[n] \). Following [31], they are modeled as
\[
v_S[n] \sim \mathcal{CN}(0, \delta_S \text{diag } \{ E\{\hat{s}[n]\hat{s}^H[n]\} \})
\] (18)
\[
v_R[n] \sim \mathcal{CN}(0, \delta_R \text{diag } \{ E\{\hat{r}[n]\hat{r}^H[n]\} \})
\] (19)
with \( 0 \leq \{ \delta_S, \delta_R \} \ll 1 \). Vectors \( w_D[n] \in \mathbb{C}^{N_D} \) and \( w_R[n] \in \mathbb{C}^{N_k} \) model limited dynamic range distortion at R and D, and are
assumed normally distributed and statistically independent of received signals at R and D, with

\[
\begin{align*}
\mathbf{w}_R[n] & \sim \mathcal{CN}(0, \mathbf{\epsilon}_R \text{diag}\{\mathbb{E}\{\mathbf{\tilde{f}}[n]\mathbf{\tilde{f}}^H[n]\}\}) \\
\mathbf{w}_D[n] & \sim \mathcal{CN}(0, \mathbf{\epsilon}_D \text{diag}\{\mathbb{E}\{\mathbf{\tilde{d}}[n]\mathbf{\tilde{d}}^H[n]\}\})
\end{align*}
\]

(20)

(21)

where \(\mathbf{\tilde{f}}[n]\) and \(\mathbf{\tilde{d}}[n]\) are the received signal prior to digital conversion at R and D, and \(0 \leq \{\mathbf{\epsilon}_R, \mathbf{\epsilon}_D\} \ll 1\). The independent Gaussian distortion models (18)-(19) and (20)-(21) accurately captures the combined effect of DAC and ADC nonlinearities and practical hardware impairments, see [31] and the references therein.

III. MMSE DESIGN

The end-to-end performance depends primarily on the ability of the system to reconstruct signal \(s[n]\) at destination. Therefore, a reasonable approach for designing filters \(\mathbf{g}_S[n], \mathbf{G}_R[n]\) and \(\mathbf{g}_D[n]\) is to minimize the MSE at node D, i.e.,

\[
\begin{align*}
\min_{\{\mathbf{g}_S[n], \mathbf{G}_R[n], \mathbf{g}_D[n]\}} \quad & \mathbb{E}\{\|\mathbf{\tilde{d}}[n] - s[n - \tau]\|^2\} \\
\text{subject to} & \quad \mathbb{E}\{\|\mathbf{s}[n]\|^2\} \leq P_S \quad \mathbb{E}\{\|\mathbf{\tilde{f}}[n]\|^2\} \leq P_R
\end{align*}
\]

(22)

where the end-to-end delay \(\tau \geq 0\) is a design parameter, and constants \(P_S > 0\) and \(P_R > 0\) denote the maximum transmit power at S and R, respectively. In view of (10), the recursive nature of the SI-affected relay makes problem (22) intractable.

To overcome this, we introduce additional linear constraints. Since SI is the dominant source of distortion in a FD relay link, see, e.g., [44], we modify problem (22) to incorporate an explicit SI suppression constraint, i.e.,

\[
\begin{align*}
\min_{\{\mathbf{g}_S[n], \mathbf{G}_R[n], \mathbf{g}_D[n]\}} \quad & \mathbb{E}\{\|\mathbf{\tilde{d}}[n] - s[n - \tau]\|^2\} \\
\text{subject to} & \quad \mathbb{E}\{\|\mathbf{s}[n]\|^2\} \leq P_S \quad \mathbb{E}\{\|\mathbf{\tilde{f}}[n]\|^2\} \leq P_R \quad \mathbf{H}_{RR}[n] \star \mathbf{G}_R[n] = 0
\end{align*}
\]

(23)

The immediate effect of introducing the linear constraints \(\mathbf{H}_{RR}[n] \star \mathbf{G}_R[n] = 0\) in (23) is summarized as follows.

Remark 1. From (10) we see that the equivalent relay impulse response has now finite order, i.e., \(\mathbf{G}_{eq}[n] = \mathbf{G}_R[n]\) and \(L_{eq} = \max\{L_{SD}, L_{SR} + L_R + L_{RD}\}\).

Remark 2. The SI is suppressed, i.e., \(\mathbf{I}[n] = 0\), making problem (23) tractable.

Remark 3. Since the SI suppression constraints constitute a linear subspace, as will be shown in Sec. IV, the available degrees of freedom in \(\mathbf{G}_R[n]\) are reduced. Therefore, problems (23) and (22) are not equivalent and their solutions may perform differently.

An important consequence of the SI suppression constraint in (23) is exposed by the next result.

Lemma 1. When \(\mathbf{H}_{RR}[n] \star \mathbf{G}_R[n] = 0\), the inverse system \((\mathbf{I}_{M_R} \delta[n] - \mathbf{G}_R[n] \star \mathbf{H}_{RR}[n])^{-1} = \mathbf{I}_{M_R} \delta[n] + \mathbf{G}_R[n] \star \mathbf{H}_{RR}[n]\) and is of finite order \(L_R + L_{RR}\).

Proof. By taking the Fourier transform of the inverse system, denoted by \(\mathcal{F}\{\mathbf{I}_{M_R} \delta[n] - \mathbf{G}_R[n] \star \mathbf{H}_{RR}[n]\}^{-1}\), we obtain

\[
\mathcal{F}\{\mathbf{I}_{M_R} \delta[n] - \mathbf{G}_R[n] \star \mathbf{H}_{RR}[n]\}^{-1} = (\mathbf{I}_{M_R} - \mathbf{G}_R(e^{j\omega})\mathbf{H}_{RR}(e^{j\omega}))^{-1}
\]

(24)

Using the Woodbury matrix identity, the right-hand side of (24) is expanded as

\[
(\mathbf{I}_{M_R} - \mathbf{G}_R(e^{j\omega})\mathbf{H}_{RR}(e^{j\omega}))^{-1} = \mathbf{I}_{M_R} + \mathbf{G}_R(e^{j\omega})(\mathbf{I}_{N_k} + \mathbf{H}_{RR}(e^{j\omega})\mathbf{G}_R(e^{j\omega}))^{-1}\mathbf{H}_{RR}(e^{j\omega})
\]

(25)

where we have used \(\mathbf{H}_{RR}(e^{j\omega})\mathbf{G}_R(e^{j\omega}) = 0\). By taking the inverse Fourier transform of (25), the inverse system impulse response is given by

\[
\mathcal{F}^{-1}\{\mathbf{I}_{M_R} + \mathbf{G}_R(e^{j\omega})\mathbf{H}_{RR}(e^{j\omega})\} = \mathbf{I}_{M_R} \delta[n] + \mathbf{G}_R[n] \star \mathbf{H}_{RR}[n]
\]

(26)

whose order is \(L_R + L_{RR}\).
By virtue of Lemma 1, the noise term \( z_R[n] \) at the relay output (15) can be written as

\[
z_R[n] = G_R[n] \ast (H_{SR}[n] \ast v_S[n] + n_R[n] + w_R[n]) + (I_{M_R} \delta[n] + G_R[n] \ast H_{RR}[n]) \ast v_R[n]
\]  

(27)

We make the following assumption about the limited dynamic range

**Assumption 1.** The statistics of \( w_R[n], w_D[n] \) and \( v_R[n] \) depend only on their respective data signal, which follows from \( \delta_j \approx 0, \delta_j \epsilon_j \approx 0, \delta_j \epsilon_j \sigma_j^2 \approx 0, \epsilon_j \delta_j \approx 0, \epsilon_j \sigma_j^2 \approx 0 \) and \( \epsilon_j \epsilon_j \approx 0 \). Let \( s_R[n] = H_{SR}[n] \ast g_S[n] \ast s[n] \) and \( s_D[n] = H_{eq}[n] \ast g_D[n] \ast s[n] \) denote the data signal arriving at \( S \) and \( D \), respectively. Therefore,

\[
w_R[n] \sim CN\left(0, \epsilon_R \text{diag} \left\{ E\{ s_R[n] \star s^H_R[n]\} \right\}\right)
\]

(28)

\[
w_D[n] \sim CN\left(0, \epsilon_D \text{diag} \left\{ E\{ s_D[n] \star s^H_D[n]\} \right\}\right)
\]

(29)

\[
v_R[n] \sim CN\left(0, \delta_R \text{diag} \left\{ E\{ G_R[n] \star s_R[n]\} \ast (G_R[n] \ast s_R[n])^H \right\}\right)
\]

(30)

Problem (23) is non-convex due to the coupling between \( g_S[n], G_R[n] \) and \( g_D[n] \). Furthermore, the covariance of \( v_S[n], w_R[n], v_R[n], w_D[n] \) are functions of filters \( g_S[n], G_R[n] \) and \( g_D[n] \), see (18)–(21). In the next section we make use of an alternating procedure to solve (23).

**IV. ALTERNATING FILTER DESIGN**

A sub-optimal solution to the non-convex problem (23) can be iteratively obtained by means of alternating (or cyclic) minimization, in which the design of each individual filter is decoupled from the others by fixing them at each iteration. The steps of such procedure are summarized in Algorithm 1. The destination node is in charge of the computation and broadcasts the filter coefficients through the feedback channel.

Symbols \( g_S^{(k)}[n], G_R^{(k)}[n] \) and \( g_D^{(k)}[n] \) denote the node filters at iteration \( k \). In summary, Algorithm 1 attempts to solve (23) by iterating over a sequence of simpler optimization problems, which, as explained in the following, are convex in their respective variables and can be solved semi-analytically. Although a global optimizer of (23) is not guaranteed, the convergence of Algorithm 1 is ensured under the conditions described in [45]. Specifically, since the sequence of MSE values obtained by the algorithm is nonincreasing and bounded below by 0, it must eventually converge. In [45], it is shown that algorithms of the same form as Algorithm 1 have linear convergence, i.e., the ratio of the norm of the difference between the current iteration and a solution and the norm of the difference between the previous iteration and a solution is less than or equal to a constant. The algorithm can be sensitive to the initialization point due to existence of several local minima. Therefore, running the algorithm with different initialization points can robustify the solution in case the algorithm falls in a local minima.

From now on, unless otherwise stated, we will simply denote \( g_S[n] = g_S^{(k)}[n], G_R[n] = G_R^{(k)}[n] \) and \( g_D[n] = g_D^{(k)}[n] \).

**A. Solution of problem (23) with respect to \( g_D[n] \)**

We first solve the problem (23) with respect to \( g_D[n] \), while keeping both \( g_S[n] \) and \( G_R[n] \) fixed, i.e.,

\[
g_D[n] = \arg \min_{g_D[n]} \mathbb{E}\{ |d[n] - s[n - \tau]|^2 \}
\]

(31)
Note that, since $g_D[n]$ is at the receive side of $D$, its optimal expression will correspond to that of the MIMO normal equations. Let $g_D$ and $g_S$ be the column-expansions of $g_D[n]$ and $g_S[n]$, respectively. From (12), we can now express $\hat{d}[n]$ as

$$\hat{d}[n] = g_D^H (\mathcal{H}_e) \mathcal{R}(g_S)s + z_D$$

(32)

where vectors $s$ and $z_D$ are defined as

$$s = [s[n] \ s[n-1] \ldots s[n-L_S-L_{eq}-L_D]^T$$

$$z_D = [z_D^n[n] z_D^n[n-1] \ldots z_D^n[n-L_D]^T$$

(33)

(34)

Let $e_\tau$ denote the $\tau$th canonical basis vector with $e_\tau[\tau] = 1$ and $0$ elsewhere. Using (32), the MSE can be expressed as

$$\mathbb{E}\{[\hat{d}[n] - s[n-\tau]^2\} = (g_D^H \mathcal{R}(\mathcal{H}_e) \mathcal{R}(g_S) - e_\tau^H) \mathcal{R}_s (\mathcal{R}^H(g_S) \mathcal{R}(\mathcal{H}_e) g_D - e_\tau) + g_D^H \mathcal{R}_s e_\tau$$

(35)

with $\mathcal{R}_s = \mathbb{E}\{ss^H\}$ and $\mathcal{R}_{z_0} = \mathbb{E}\{z_Dz_D^H\}$ (see Table I for a full expression) denoting the autocorrelation matrices of $s[n]$ and $z_D[n]$, respectively. Combining (35) and (31), the optimal filter $g_D^*\$ is given by

$$g_D^* = \mathcal{R}_D^{-1} \mathcal{R}(\mathcal{H}_e) \mathcal{R}(g_S) \mathcal{R}_s e_\tau$$

(36)

where

$$\mathcal{R}_D = \mathcal{R}(\mathcal{H}_e) \mathcal{R}(g_S) \mathcal{R}^H(g_S) \mathcal{R}(\mathcal{H}_e) + \mathcal{R}_{z_0}$$

(37)

is the autocorrelation matrix of the input signal to $g_D[n]$. From (36) we see that $\tau$, with $0 \leq \tau \leq (L_{eq} + L_D + L_S)$, selects one of the columns of $\mathcal{R}_D^{-1} \mathcal{R}(\mathcal{H}_e) \mathcal{R}(g_S) \mathcal{R}_s$, and can be interpreted as a filter selector.

### B. Solution of problem (23) with respect to $G_R[n]$

The next step of Algorithm 1 is to design $G_R[n]$ as the solution to (23) assuming both $g_S[n]$ and $g_D[n]$ fixed, i.e.,

$$G_R[n] = \arg \min_{G_S[n]} \mathbb{E}\{[\hat{d}[n] - s[n-\tau]^2\}$$

subject to

$$\mathbb{E}\{[\hat{r}[n]^2\} \leq P_R$$

$$H_{RR}[n] * G_R[n] = 0$$

(38)

Even under the SI suppression constraints, solving (38) requires the use of certain approximations. Concretely, from (27), the power of $z_R[n]$ has the following expression

$$\mathbb{E}\{[z_R[n]^2\} = \mathbb{E}\{[G_R[n] * n_R[n]^2\} + \mathbb{E}\{[v_R[n]^2\}$$

$$+ \mathbb{E}\{[G_R[n] * H_{SR}[n] * v_S[n]^2\}$$

$$+ \mathbb{E}\{[G_R[n] * w_R[n]^2\}$$

$$+ \mathbb{E}\{[G_R[n] * H_{RR}[n] * v_R[n]^2\}$$

(39)

From (7) and (19) we see that the covariance of $v_R[n]$ depends on $G_R[n]$. Consequently, the last term in (39), $\mathbb{E}\{[G_R[n] * H_{RR}[n] * v_R[n]^2\}$, exhibits a “fourth-order” relation with $G_R[n]$, as seen in (40).

To avoid such higher-order relation, we approximate $G_R^{(k)}[n] * H_{RR}[n]$ as $G_R^{(k-1)}[n] * H_{RR}[n]$. This transforms (39) into

$$\mathbb{E}\{[z_R[n]^2\} = \mathbb{E}\{[G_R^{(k)}[n] * n_R[n]^2\} + \mathbb{E}\{[v_R[n]^2\}$$

$$+ \mathbb{E}\{[G_R^{(k)}[n] * H_{SR}[n] * v_S[n]^2\}$$

$$+ \mathbb{E}\{[G_R^{(k)}[n] * w_R[n]^2\}$$

$$+ \mathbb{E}\{[G_R^{(k-1)}[n] * H_{RR}[n] * v_R[n]^2\}$$

(41)

with

$$\mathbb{E}\{[v_R[n]^2\} = \delta_R \text{tr}\{\text{diag}\{g_R^{(k)}R_{sr}(g_R^{(k)})^H\}\}$$

(42)

Matrix $R_{sr} = \mathbb{E}\{s_Rs_R^H\}$ is the autocorrelation of $s_R[n] = H_{SR}[n] * g_S[n] * s[n]$, see (30). Vector $s_R$ is defined as

$$s_R = [s_R^n[n] \ s_R^n[n-1] \ldots s_R^n[n-L_R]^T$$

(43)

By using the approximation in (41), $\mathbb{E}\{[z_R[n]^2\}$ has a second-order relation with $G_R^{(k)}[n]$, and (38) can be cast as a constrained least-squares problem. The difference between (39) and (41) is roughly proportional to $\delta_R$ and the approximation
Algorithm 2 Design steps for $G_R^{(k)}[n]$

1: Calculate the nullspace basis $N$ of $(I_{N_R} \otimes C(\mathcal{H}_{RR}))$.
2: Approximate $\mathbb{E}\{||\hat{z}_R[n]||^2\}$ according to expression (41).
3: Calculate $w^*$ by solving (49).
4: Obtain $G_R^{(k)}[n]$ through expression $g_R^* = Nw^*$.

in (41) will be tight whenever $\delta$ is not the dominant noise source. Typically, noise due to limited dynamic range is lower than thermal noise [31].

Although the SI suppression constraints in (38), $H_{RR}[n] \ast G_R[n] = 0$, force the possible solutions to lie within a subspace of reduced dimension, (38) cannot be analytically solved by means of the normal equations due to the second-order power inequality constraints $\mathbb{E}\{||\hat{r}[n]||^2\} \leq P_R$. To characterize the interference-free subspace, we arrange constraints $H_{RR}[n] \ast G_R[n] = 0$ in the following matrix form

$$C(\mathcal{H}_{RR}) \begin{bmatrix} G^T_R[0] & \ldots & G^T_R[L_R] \end{bmatrix}^T = \begin{bmatrix} 0 & \ldots & 0 \end{bmatrix}^T$$

Applying the vec operator, (44) can be written in a compact form

$$(I_{N_R} \otimes C(\mathcal{H}_{RR}))g_R = 0$$

with $g_R = \text{vec}\{[G^T_R[0] \ldots G^T_R[L_R]]^T\} \in \mathbb{C}^{N_R M_R(L_R+1) \times 1}$. For a nontrivial solution to (44) to exist, the rank of $C(\mathcal{H}_{RR})$ must be strictly less than the number of its columns, which is $M_R(L_R + 1)$. Assuming a full-row rank $C(\mathcal{H}_{RR})$, this condition reads as $N_R(L_R + L_{RR}) < M_R(L_R + 1)$. For this to hold, the number of transmit antennas must exceed the number of receive antennas ($N_R > L_{RR}$), and in addition, the order of the relay filter must satisfy $L_R \geq N_R (L_{RR} - 1)/(M_R - N_R)$. Consequently, the subspace of feasible $G_R[n]$ is given by

$$g_R = Nw$$

where the columns of $N$ constitute a basis of the nullspace of $(I_{N_R} \otimes C(\mathcal{H}_{RR}))$ and $w$ is an arbitrary vector. Due to the constraints (45), the number of degrees of freedom (DoF) in $G_R[n]$ is equal to the dimension of this null space, which is given by

$$\text{DoF} = N_R (M_R(L_R + 1) - \text{rank}\{C(\mathcal{H}_{RR})\})$$

$$= N_R M_R(L_R + 1) \rho$$

where $\rho$ defines the fraction of DoF

$$\rho = 1 - \frac{\text{rank}\{C(\mathcal{H}_{RR})\}}{M_R(L_R + 1)} \in [0, 1]$$

When $\rho \to 1$ either $H_{RR}[n] \to 0$ or $M_R(L_R + 1)$ is very large, whereas when $\rho \to 0$ the degrees of freedom in the relay that are used to suppress the self-interference is maximized. For a given relay configuration, $\rho$ is minimized when $C(\mathcal{H}_{RR})$ is of full-rank. Generally speaking, $C(\mathcal{H}_{RR})$ is of full-rank when $H_{RR}[n]$ follows a Rayleigh fading channel model, setting a worst case scenario benchmark in terms of self-interference.

Combining (46) and (38) yields the equivalent optimization problem

$$w^* = \arg \min_w \mathbb{E}\{||\hat{d}[n] - s[n - \tau]||^2\}$$

subject to $\mathbb{E}\{||\hat{r}[n]||^2\} \leq P_R$ (49)

Problem (49) is a Least Squares minimization problem with a Quadratic Inequality constraint (LSQI) [46]. Appendix A provides the details of the computation and a semi-analytical expression of the solution. Finally, the optimal $G_R^*[n]$ is recovered directly from $g_R^* = Nw^*$ and reordering of $g_R = \text{vec}\{[G^T_R[0] \ldots G^T_R[L_R]]^T\}$. Table 2 summarizes the steps needed to design filter $G_R[n]$ at each iteration.

C. Solution of problem (23) with respect to $g_S[n]$

In order to complete an iteration, and as a final step, we must solve (23) with respect to $g_S[n]$ for both $G_R[n]$ and $g_R[n]$ fixed, i.e.,

$$g_S^*[n] = \arg \min_{g_S[n]} \mathbb{E}\{||\hat{d}[n] - s[n - \tau]||^2\}$$

subject to $\mathbb{E}\{||\hat{r}[n]||^2\} \leq P_S$ (50)

Since $g_S[n]$ is located at node S, its optimal expression will be that of a precoding filter tailored to the combined channel $g_D[n] \ast H_{eq}[n]$. Similarly to previous step, problem (50) can be cast as an LSQI problem and solved using the theory in [46]. We refer the reader to Appendix B for details of the computation and a semi-analytical expression of the solution.
V. Simulation Results and Discussion

We respectively define the signal-to-noise ratio of the SR hop, the SD hop and the RD hop as

\[
\text{SNR}_{SR} = \frac{\mathbb{E}\{||H_{SR}[n] * g_{S}[n] * s[n]||^2\}}{\mathbb{E}\{||n_{R}[n]||^2\}}
\]

\[
\text{SNR}_{SD} = \frac{\mathbb{E}\{||H_{SD}[n] * g_{S}[n] * s[n]||^2\}}{\mathbb{E}\{||n_{D}[n]||^2\}}
\]

\[
\text{SNR}_{RD} = \frac{\mathbb{E}\{||H_{RD}[n] * g_{R}[n] * H_{SR}[n] * g_{S}[n] * s[n]||^2\}}{\mathbb{E}\{||n_{D}[n]||^2\}}
\]

The final signal-to-noise ratio at the destination takes into account the limited dynamic range parameters \(\delta_{S}, \delta_{R}, \epsilon_{R} \text{ and } \epsilon_{D}\) and is defined as

\[
\text{SNR}_{D} = \frac{\mathbb{E}\{||g_{S}[n]^{H} * H_{eq}[n] * g_{S}[n] * s[n]||^2\}}{\mathbb{E}\{||g_{S}[n]^{H} * z_{D}[n]||^2\}}
\]

We consider a source node with \(M_{S} = 2\) antennas transmitting a 64-QAM OFDM modulated signal. The number of subcarriers is \(n_{c} = 1024\) and the cyclic prefix length is \(n_{p} = 32\). The sampling frequency equals the Nyquist frequency and \(\mathbb{E}\{|s[n]|^2\} = 1\). Destination node has \(N_{D} = 2\) receive antennas, while the relay node has \(N_{R} = 2\) receive antennas and \(M_{R} = 5\) transmit antennas. The maximum transmit powers at the source and the relay are set to \(P_{S} = 0\ dB\) and \(P_{R} = 0\ dB\), respectively. The channels follow a Rayleigh fading model, so each tap of \(H_{ij}[k]\) is independently drawn from a circularly-symmetric complex normal distribution, i.e., \(\text{vec}\{H_{ij}[k]\} \sim \mathcal{CN}(0, I_{N_{j}M_{i}})\) and then a scaling is performed in order to set \(\|H_{ij}\|^2 = \gamma_{ij}\). Parameter \(\gamma_{ij}\) modifies the SNR for a fixed noise power \(\sigma_{n}^2\). In this work, we only consider perfect channel state information, therefore, the following results can be seen as a benchmark case. The direct link is weaker than the relay hop and subjected to stronger multipath components, and channel orders are \(L_{SD} = 5\), \(L_{SR} = 2\), \(L_{RD} = 2\) and \(L_{RR} = 2\) and normalization constants are \(\gamma_{SD} = 0.1\), \(\gamma_{SR} = 1\), \(\gamma_{RD} = 1\) and \(\gamma_{RR} = 1\), respectively. Noise levels are \(\delta_{S} = \delta_{R} = -30\ dB\), \(\epsilon_{R} = \epsilon_{D} = -20\ dB\), and \(\sigma_{n}^2 = \sigma_{D}^2 = -20\ dB\). Filters \(g_{S}[n]\), \(g_{R}[n]\) and \(g_{D}[n]\) all have orders \(L_{S} = L_{R} = 2\). Consequently, from (48) we have that a fraction DoF \(\rho = 0.5\) is used by \(g_{R}[n]\) for suppressing the self-interference. The end-to-end delay is set to \(\tau = 3\) by using the rule of thumb recommending a value half the length of the equivalent channel, \(L_{eq} = 7\). A method to optimize \(\tau\) can be found in [47], [48], and a study about its influence on the performance in [49].

Algorithm 1 is assumed to have converged when the weighted filter norm between two consecutive iterations falls below a certain threshold \(\varepsilon^2 \geq 0\), i.e.,

\[
\|g_{S}[k] - g_{S}[k-1]\|^2/\|g_{S}[k]\|^2 \leq \varepsilon^2
\]

\[
\|G_{R}[k] - G_{R}[k-1]\|^2/\|G_{R}[k]\|^2 \leq \varepsilon^2
\]

\[
\|g_{D}[k] - g_{D}[k-1]\|^2/\|g_{D}[k]\|^2 \leq \varepsilon^2
\]

where the sensitivity threshold \(\varepsilon\) is set to \(\varepsilon^2 = 0.003\). The number of iterations to reach convergence varies as function of the system parameters and channel impulse responses. From the observed results, the algorithm may take from a few dozens up to several thousands iterations.

Initialization points are \(g_{S}^{(0)}[n] = 1\delta[n], g_{D}^{(0)}[n] = 1\delta[n]\) and \(w^{(0)} = 1\), with \(1\) being the all-ones vector of appropriate dimension. The results are obtained by averaging over independent realizations. The parameters listed above are fixed throughout simulations unless otherwise stated.

Figure 2 depicts the MSE as a function of \(\text{SNR}_{SD}\) and \(\text{SNR}_{RD}\), when \(\gamma_{SR} = 1\), \(\gamma_{RD} \in [0.1, 1]\) and \(\gamma_{SD} \in [0, 1]\). We see that for low values of \(\text{SNR}_{RD}\) the presence of the direct link, i.e., when \(\text{SNR}_{SD} > -\infty\), improves the end-to-end performance by approximately 5 dB. On the other hand, when \(\text{SNR}_{RD}\) is large, the improvement due to the direct link is roughly 1 dB. We can conclude that the contribution of the direct link to the overall performance is a nonlinear function of the SNR of the RD hop. When the RD hop supports a high SNR, the contribution of the direct link to the end-to-end performance is marginal.

Figure 3 shows the MSE as a function of \(\text{SNR}_{SR}\) and \(\text{SNR}_{RD}\) when \(\text{SNR}_{RD} = 0\), i.e., no direct link, \(\gamma_{SR} \in [0.1, 1]\) and \(\gamma_{RD} \in [0.1, 1]\). We see that the impact of \(\text{SNR}_{SR}\) on the MSE is more significant than that of \(\text{SNR}_{RD}\), due to the noise from S being propagated to D. For low \(\text{SNR}_{SR}\), increasing \(\text{SNR}_{RD}\) by 10 dB results in an MSE improvement of 3 dB, whereas, when \(\text{SNR}_{SR}\) is large, the same operation results in an MSE improvement of 5.5 dB. Therefore, a system where the SR hop supports higher SNR values than the RD hop leads to better performance.

Figure 4 depicts the MSE as a function of \(\text{SNR}_{SR}\) and \(\text{SNR}_{RD}\), when \(\gamma_{RD} = 1\), \(\gamma_{SR} \in [0.1, 1]\) and \(\gamma_{SD} \in [0, 1]\). Similarly to the results in Fig. 2, the MSE gain decreases as \(\text{SNR}_{SR}\) increases.

Figure 5 shows the contour lines of the MSE as a function of the distortion at the destination, \(\epsilon_{D}\), and the distortion at the relay, \(\epsilon_{R}\). From the obtained results, although the MSE depends on both \(\epsilon_{D}\) and \(\epsilon_{R}\), \(\epsilon_{D}\) has a stronger impact on the end-to-end
Fig. 2. MSE versus SNR_{RD} for various SNR_{SD}.

Fig. 3. MSE versus SNR_{RD} for various SNR_{SR}. No S-D link.

performance than ε_R, e.g., compare the point (−10,−30) with the point (−30,−10) in Fig. 5. This is a consequence of the cumulative effect of w_D[n] at D, whose power depends on other noise sources, such as w_R[n].

Figure 6 shows the contour lines of the MSE as a function of the source transmit noise, δ_S, and the relay transmit noise, δ_R. Note the similar behavior of Fig. 6 and Fig. 5. It is observed that δ_S, or δ_R, has a stronger impact on the MSE than δ_S. For instance, comparing the point (δ_R = −10, δ_S = −30) and the point (δ_R = −30, δ_S = −10) of Fig. 6, it is seen that there is approximately a 1 dB gap between both. This is a consequence of the relay transmit noise being filtered by H_{RR}[n].

Figure 7 shows the final SNR at destination SNR_D for different values of SNR_{RD} and SNR_{SD}. The case SNR_{RD} → −∞ indicates the absence of a relay. Therefore, Fig. 7 shows the performance gain resulting from the relay in the link. Note the saturation effect when SNR_{SD} is large, where performance gain is 3−4 dB. On the other hand, when SNR_{SD} is low, the relay can boost performance in 10−15 dB.

Figure 8 compares the final SNR at destination SNR_D for different values of SNR_{SR} and SNR_{SD}. Solid lines depict the unlimited dynamic range (u.d.r.) cases, δ_S = ε_R = δ_R = ε_D = 0, whereas dashed lines depict the limited dynamic range (l.d.r.) cases, δ_S = δ_R = −30 dB and ε_R = ε_D = −20 dB. The gap between l.d.r and u.d.r. cases is of approximately 1−2 dB, and it increases alongside SNR_{SD}.

Figure 9 shows the MSE as a function of the fraction of DoF ρ in (48) for different values of M_R and L_R. Dashed lines show ρ for the same values of M_R and L_R. Note the clear relation between ρ and the MSE. Figure 9 highlights that different values of M_R and L_R may lead to the same ρ or MSE. This is relevant if additional coefficients of G_R[n] are affordable, as performance may reach that of a system with more antennas, for example, cases (M_R = 7, L_R = 10) and (M_R = 9, L_R = 1) results in the same MSE. However, M_R has a bigger impact on the performance than L_R.

Figure 9 also highlights the difference in performance between an AF relay (L_R = 0) and an FF relay. In all the tested cases, the FF relay outperforms the AF relay by 2−3 dB. In fact, an FF relay with less antennas may outperform an AF relay
equipped with more antennas. This is, because for an AF relay, $\rho$ is low even if $M_R$ is large. Consequently, an FF protocol has clearly the edge in performance over an AF protocol, particularly in the case where the number of transmit antennas $M_R$ is fixed and $L_R$ is set on demand.

Figures 10 and 11 show the histogram of the number of iterations and the MSE for randomized initialization points when simulations parameters are fixed. The number of randomized initialization points is 1000, whereas simulations parameters are set to default values. From Fig. 10 the number of iterations to reach convergence is of the order of several dozens. Concretely, most of the times the algorithm takes than 25 iterations before obtaining a solution. Note that this number depends on the initialization point and the value of $\varepsilon_2$, where lower values of $\varepsilon_2$ result, on average, on a higher number of iterations. Similarly, because the stop criterion is based on the normalized change per iteration, the resulting MSE follows the histogram in Fig. 11. Note that the MSE is most likely to lie within an interval of $1.5 - 2$ dB around its mean value.

VI. CONCLUSIONS

We presented a method for MMSE design of node filters in a filter-and-forward full-duplex MIMO relay link subjected to limited dynamic range. The original non-convex optimization problem is approximated by an alternating optimization algorithm, where each node’s filter is designed individually at each iteration. Simulations show that the balance between direct path (source-destination) and relay path (source-relay-destination) has a strong influence on the end-to-end performance, particularly when the source-to-relay path supports low SNR. A filter-and-forward protocol outperforms an amplify-and-forward protocol in most of the cases, even for fewer antennas. Limited dynamic range decreases the performance by a factor that depends on the individual hop SNR.
Fig. 6. MSE contour lines (in dB) versus source transmit noise $\delta_S$ and relay transmit noise $\delta_R$.

Fig. 7. SNR$_D$ versus SNR$_{SD}$ for various SNR$_{RD}$.

**APPENDIX A**

*Solution of (23) in terms of $G_R[n]$ for fixed $g_S[n]$ and $g_D[n]$. Let first write the MSE as a function of $w$. Using (32), $\hat{d}[n]$ has the following expression*

$$\hat{d}[n] = g_D^H (\mathcal{R}(H_{RD}) \mathcal{R}(\mathcal{G}_R) \mathcal{R}(\mathcal{H}_{SR}) + \mathcal{R}(\mathcal{H}_{SD})) \mathcal{R}(g_S)s$$

$$+ g_D^H z_D$$

(51)

where $s$ denotes a vector of appropriate size that contains current and $L$ past samples of $s[n]$, i.e.,

$$s = [s[n] \; s[n-1] \; \ldots \; s[n-L]]^T$$

(52)

By vectorizing both sides of (51), we obtain

$$\hat{d}[n] = g_D^H (\mathcal{R}(H_{RD}) \mathcal{R}(g_S)s + z_D)$$

$$+ g_D^H (\mathcal{R}(H_{SR}) \mathcal{R}(g_S)s)^T \otimes \mathcal{R}(H_{RD})) P w$$

(53)

where $P = VN$, and $V$ being a reordering matrix satisfying $\text{vec}(\mathcal{R}(\mathcal{G}_R)) = V \text{vec}(\mathcal{G}_R)$, see Appendix C. From (53),

$$\mathbb{E}\{ |\hat{d}[n] - s[n-\tau]|^2 \} = w^H Q_s w - 2 \text{Re}\{ w^H q_{\tau} \} + g_D^H (\mathcal{R}(H_{RD}) \mathcal{R}(g_S)s)^T \otimes \mathcal{R}(H_{RD})) P w$$

(54)

where matrix $Q_s$, vector $q_{\tau}$ and scalar $q$ do not depend on $w$ and are defined in Table II. Matrix $R_{w_0}$ denotes the autocorrelation matrix of $w_D[n]$ that depends on $G_R[n]$. 

Equation (55) expresses $g_D^H R_{w_0} g_D$ as a convex function of $w$, where in Equality 1 we used the linearity of expectation and the properties of Kronecker product, whereas in Equality 2 we used Lemmas 2 and 3, presented below.

**Lemma 2.** For any constant matrices $A \in \mathbb{C}^{M \times N}$ and $C \in \mathbb{C}^{P \times R}$ and any random matrix $B \in \mathbb{C}^{R \times M}$, the following property holds

$$\mathbb{E}\{\text{tr}\{C \text{diag}\{B A A^H B^H\} C^H}\}\} = \text{tr}\{A^H \mathbb{E}\{B^H \text{diag}\{C^H C\} B\} A\} \quad (56)$$

**Proof.** Let $e_i \in \mathbb{C}^R$ denote the $i$th canonical basis vector. Matrix $J_i = e_i e_i^T \in \mathbb{C}^{R \times R}$ denotes the single-entry matrix whose $i$th diagonal element is one and zero elsewhere. The diag operation can be written as

$$\text{diag}\{B A A^H B^H\} = \sum_{i=1}^{R} J_i B A A^H B^H J_i^H \quad (57)$$

Combining (57) and (56) yields

$$\mathbb{E}\{\text{tr}\{C \text{diag}\{B A A^H B^H\} C^H\}\} = \mathbb{E}\{\text{tr}\{C \sum_{\rho=1}^{R} J_{\rho} B A A^H B^H J_{\rho}^H C^H\}\}$$
Fig. 10. Histogram of the number of iterations until convergence for randomized initialization points.

Fig. 11. Histogram of the MSE in dB for randomized initialization points.

\[ g_D^H R_{w0} g_D = g_D^H \left( I_{L_D+1} \otimes \varphi D \right) \text{diag} \left\{ \left( R(\mathcal{H}_{RD}) R(G_R) R(\mathcal{H}_{SR}) R(g_s) \right) \mathbb{E}\{ s s^H \} \left( R(\mathcal{H}_{RD}) R(G_R) R(\mathcal{H}_{SR}) R(g_s) \right)^H \right\} g_D \]

\[ = \epsilon_D \sum_{i=0}^{L_D} \mathbb{E}\{ \varphi_D^H[i] \text{diag} \left\{ \left( R(\mathcal{H}_{SR}) R(g_s) s \right)^T \otimes R(\mathcal{H}_{RD}) \right\} P_{ww}^H P^H \left( \left( R(\mathcal{H}_{SR}) R(g_s) s \right)^T \otimes R(\mathcal{H}_{RD}) \right)^H \varphi_D[i] \} \]

\[ = \epsilon_D w^H \mathbb{E}\left\{ P^H \left( R(\mathcal{H}_{SR}) R(g_s) s \right)^T \otimes R(\mathcal{H}_{RD}) \right)^H I_{N_0[L_D]} \text{diag} \left\{ g_D^H g_D^H \right\} I_{N_0[L_D]} \left( R(\mathcal{H}_{SR}) R(g_s) s \right)^T \otimes R(\mathcal{H}_{RD}) P \right\} w \]  

(55)

\[ = \sum_{p=1}^{R} \mathbb{E}\{ \text{tr}\{ A^H B^H J_p^H C^H C J_p^H B A \} \} \]

\[ = \text{tr}\{ A^H \mathbb{E}\{ B^H \sum_{p=1}^{R} (J_p^H C^H C J_p^H) B \} A \} \]

\[ = \text{tr}\{ A^H \mathbb{E}\{ B^H \text{diag}(C^H C) B \} A \} \]  

(58)

where we have used that $\text{tr}\{ A A^H \} = \text{tr}\{ A^H A \}$ and that $J_i$ is a Hermitian matrix.
Lemma 3. For any filter $A[n] \in \mathbb{C}^{M \times N}$ of order $L_A$ the following property holds

$$\sum_{k=0}^{L_A} \text{diag} \{A^H[k]A[k]\} = I_N[L_A] \text{diag} \{A^H[A]\} I_N[L_A]$$

(59)

Proof. Gramian matrix $A^H A$ has the following expression


(60)

which, after applying the diag operation, results in

$$\text{diag} \{A^H A\} = \begin{pmatrix} \text{diag} \{A^H[0]A[0]\} \\ \ddots \\ \text{diag} \{A^H[L_A]A[L_A]\} \end{pmatrix}$$

(61)

from where the result of the lemma follows immediately. □

Proceeding in a similar fashion with the additional terms in $g_D^H \mathcal{R}(\mathcal{H}_{SD}) R_{\tau_0} \mathcal{R}^H(\mathcal{H}_{SD}) g_D$, the MSE can be written as

$$\mathbb{E}\{|\hat{d}[n] - s[n - \tau]|^2\} = w^H Q w - 2 \text{Re}\{w^H q_D\} + q$$

(62)

where $Q = Q_s + Q_{\tau_0} + Q_{\tau} + \delta_s + \delta_{\tau_0} + \delta_{\tau}$. Each matrix $Q_i$ is a result of noise source with parameter $i$ and signal $s[n]$ propagating to D and is defined in Table II.

The inequality constraint, $\mathbb{E}\{|\hat{f}[n]|^2\} \leq P_f$, can be written as

$$\mathbb{E}\{|\hat{f}[n]|^2\} = w^H Q_r w \leq P_r$$

(63)

with $Q_r = N^H (R_r^T \otimes I_{M_k})^H N$ and $R_r$ the autocorrelation of $r[n]$, see Table II. Finally, the optimization problem can be expressed as:

$$\min_w \quad w^H Q w - 2 \text{Re}\{w^H q_r\} + q$$

subject to $w^H Q_r w \leq P_r$ (64)

Problem (64), as explained in [46], is an LSQI problem, whose unique solution is obtained as follows:

1) If the unconstrained solution $w^* = Q^{-1} q_r$ satisfies $(w^*)^H Q_r w^* \leq P_r$ then $w^*$ is a solution of (64).

2) If not, the optimal solution is obtained using the Lagrange function and is given by $w^* = (Q + \lambda^* \mathbb{R}_r)^{-1} q_r$ where $\lambda^*$ is a Lagrange multiplier satisfying $(w^*)^H Q_r w^* = P_r$.

The Lagrange multiplier $\lambda^*_r$ can be obtained by means of any standard root finding technique.

APPENDIX B

Solution of (23) in terms of $g_S[n]$ for fixed $G_R[n]$ and $G_S[n]$. Let $f_S = \text{vec}\{[g_S[0] \ldots g_S[L_S]]\} \in \mathbb{C}^{M(L_S+1)}$. By vectorizing both sides of (32), we obtain

$$\hat{d}[n] = (s^T \otimes g_D^H \mathcal{R}(\mathcal{H}_{eq})) V f_S + g_D^H z_D$$

(65)

with $V$ being matrix satisfying $\text{vec}\{\mathcal{R}(g_S)\} = V f_S$, see Appendix C. From (65),

$$\mathbb{E}\{|\hat{d}[n] - s[n - \tau]|^2\} = f_S^H T_s f_S - 2 \text{Re}\{f_S^H t_r\} + g_D^H R_{\tau_0} g_D$$

(66)

where matrix $T_s$ and vector $t_r$ are defined in Table III. Proceeding in a similar fashion as in Appendix A, term $g_D^H R_{\tau_0} g_D$ can be expressed as a convex function of $f_S$, resulting in

$$\mathbb{E}\{|\hat{d}[n] - s[n - \tau]|^2\} = f_S^H T_s f_S - 2 \text{Re}\{f_S^H t_r\} + t$$

(67)

where $T = T_s + T_{\tau_0} + T_{\tau} + T_{\delta_s} + T_{\delta_{\tau_0}} + T_{\delta_{\tau}}$ and scalar $t$ collects all the terms independent of $g_S$. Each matrix $T_i$ is a result of noise source $i$ and signal $s[n]$ propagating to D and is defined in Table III.

The inequality constraint, $\mathbb{E}\{|\hat{s}[n]|^2\} \leq P_s$, can be written in terms of $g_S$ as

$$\mathbb{E}\{|\hat{s}[n]|^2\} = f_S^H T_s f_S \leq P_s$$

(68)
with $T_s = (R_s^T \otimes I_{M_s})^H$ and $R_s$ the autocorrelation of $s[n]$. Finally, $f_s$ is the solution to the following optimization problem:

$$
\begin{align*}
\min_{f_s} & \quad f_s^H T f_s - 2 \Re \{f_s^H t_r\} + t \\
\text{subject to} & \quad f_s^H T f_s \leq P_s
\end{align*}
$$

(69)

From Appendix A, (69) is seen to be an LSQI problem, whose solution $f_s^*$ is given by

$$
f_s^* = (T + \lambda_s^* T_s)^{-1} t_r

(70)

where Lagrange multiplier $\lambda_s^* \geq 0$ ensures $(f_s^*)^H T f_s^* = P_s$ when the unconstrained solution $f_s^* = T^{-1} t_r$ yields $(f_s^*)^H T f_s^* > P_s$.

APPENDIX C

**Proof of** $\text{vec}\{R(B)\} = V \text{vec}\{B\}$. In terms of $B$, (2) can be rewritten as

$$
R(B) = \sum_{j=0}^{L_2} U_j B W_j
$$

(71)

where $U_j \in \mathbb{C}^{N(L_2+1) \times N}$ and $W_j \in \mathbb{C}^{P(L_2+1) \times P(L_2+1)}$ are given by

$$
U_j = \begin{bmatrix} 0_{P(L_2+1) \times jP} & I_{P(L_2+1)} & 0_{P(L_2+1) \times (L_2+1)P} \end{bmatrix}^T
$$

(72)

$$
W_j = \begin{bmatrix} 0_{P(L_2+1) \times jP} & I_{P(L_2+1)} & 0_{P(L_2+1) \times (L_2+1)P} \end{bmatrix}
$$

(73)

Using (71) and the properties of the vectorization operator,

$$
\text{vec}\{R(B)\} = \text{vec}\left\{ \sum_{j=0}^{L_2} U_j B W_j \right\}
$$

$$
= \sum_{j=0}^{L_2} \text{vec}\{U_j B W_j\}
$$

$$
= \sum_{j=0}^{L_2} (W_j^T \otimes U_j) \text{vec}\{B\}
$$

$$
= V \text{vec}\{B\}
$$

(74)

where the resulting vector $\text{vec}\{B\} \in \mathbb{C}^{NP(L_2+1)}$ and matrix $V = \sum_{j=0}^{L_2} (W_j^T \otimes U_j) \in \mathbb{C}^{NP(L_2+1) \times N P(L_2+1)}$.

**TABLE I**

**AUTOCORRELATION MATRIX OF $z_0[n]$, $R_{zd}$**

<table>
<thead>
<tr>
<th>$L_{RD'} = L_R + L_{RD} + L_D$</th>
<th>$L_{RD''} = L_{RD'} + L_{RR}$</th>
</tr>
</thead>
<tbody>
<tr>
<td>$R_{zd} = \delta_R R(H_{RD}) R(G_{R}) R(H_{RR}) \left( I_{L_{RD'} + 1} \right) \otimes \text{diag} \left{ G_{R} R(H_{SR}) R(g_s) R_2 R^H(g_s) R_{SR} R^H(H_{SR}) G_{R}^H \right} R^H(H_{RR}) R^H(G_{R}) R^H(H_{RD})$</td>
<td></td>
</tr>
<tr>
<td>$+ \delta_R R(H_{RD}) \left( I_{L_{RD} + L_{RD} + 1} \right) \otimes \text{diag} \left{ G_{R} R(H_{SR}) R(g_s) R_2 R^H(g_s) R_{SR} R^H(H_{SR}) G_{R}^H \right} R^H(H_{RD}) + \sigma_{NR}^2 R(H_{RD}) R(G_{R}) R^H(G_{R}) R^H(H_{RD})$</td>
<td></td>
</tr>
<tr>
<td>$+ \epsilon_{R} R(H_{RD}) R(G_{R}) \left( I_{L_{RD'} + 1} \right) \otimes \text{diag} \left{ H_{SR} R(g_s) R_2 R^H(g_s) H_{SR} \right} R^H(G_{R}) R^H(H_{RD}) + \sigma_{NR}^2 I_{N_R(L_{RD} + 1)}$</td>
<td></td>
</tr>
<tr>
<td>$+ \epsilon_{D} \left( I_{L_{D} + 1} \right) \otimes \text{diag} \left{ H_{SR} R(g_s) R_2 R^H(g_s) H_{SR} \right} + \delta_{R} R(H_{eq}) \left( I_{L_{D} + L_{eq} + 1} \right) \otimes \text{diag} \left{ g_s R_2 g_s^H \right} R^H(H_{eq})$</td>
<td></td>
</tr>
</tbody>
</table>

ACKNOWLEDGMENT

This work was partially supported by the Academy of Finland under Grants 296849 and 288249, by the Research Council of Norway, the Agencia Estatal de Investigación (Spain) and the European Regional Development Fund (ERDF) under project WINTER (TEC2016-76409-C2-2-R), and by the Xunta de Galicia (Agrupación Estratéxica Consolida de Galicia accreditation 2016-2019; Red Temática RedTEIC 2017-2018).
TABLE II
MSE IN TERMS OF $\mathbf{G}_R[n]$ FOR FIXED $\mathbf{g}_S[n]$ AND $\mathbf{g}_D[n]$. OPERATOR * DENOTES CONJUGATION.

<table>
<thead>
<tr>
<th>DEFINITIONS</th>
</tr>
</thead>
<tbody>
<tr>
<td>${L_{\text{ID}} = L_{\text{SR}} + L_{\text{AR}} + L_{\text{AB}} + L_{\text{CD}} }$, ${L_{\text{ID}} = L_{\text{SR}} + L_{\text{AR}} + L_{\text{AB}} + L_{\text{CD}} }$, ${D_k[n] = I_{D_k} \delta[n] + G_k^{(\text{ID})}[n] \cdot \mathbf{H}_{\text{SR}}[n] }$</td>
</tr>
<tr>
<td>$Q_1 = P^H \mathbb{E} \left{ (\mathbf{R}(\mathbf{H}_{\text{SR}}) \mathbf{R}(\mathbf{g}<em>k) s^* \otimes \mathbf{R}^H(\mathbf{H}</em>{\text{AR}}) \mathbf{g}_S) \right} P$</td>
</tr>
<tr>
<td>$Q_{\text{AR}} = P^H \left( (\mathbf{R}(\mathbf{H}<em>{\text{SR}}) \mathbf{R}(\mathbf{g}<em>k) \mathbf{E} { s s^* } - \mathbf{R}^H(\mathbf{H}</em>{\text{AR}}) \mathbf{g}<em>S^\dagger \mathbf{H}</em>{\text{AR}}^H)(\mathbf{H}</em>{\text{SR}}) \right) P$</td>
</tr>
</tbody>
</table>

MSE as a function of $\mathbf{w}$, $\mathbf{g}_S = \mathbf{Nw} = \text{vec} \{ \left[ \mathbf{G}_S^T[0] \ldots \mathbf{G}_S^T[L_{\text{SR}}] \right] \}$

$\mathbb{E} \left\{ |d[n] - s[n-t]^2| \right\} = \mathbf{w}^H \left( Q_{\text{AR}} + Q_{\text{SR}} + Q_{\text{ID}} \right) \mathbf{w} - 2 \text{ Re} \left\{ \mathbf{w}^H \mathbf{q}_t \right\} + q$

REFERENCES

TABLE III

MSE in terms of $g_5[n]$ for fixed $G_T[n]$ and $g_{L_3}[n]$. Operator $^*$ denotes conjugation.

<table>
<thead>
<tr>
<th>DEFINITIONS</th>
</tr>
</thead>
<tbody>
<tr>
<td>$L_{RD'} = L_R + L_{RD} + L_D$, $L_{RD''} = L_{RD'} + L_{RR}$, $D_{FF}[n] = I_{MK} s[n] + G_R[n] * H_{RR}[n]$</td>
</tr>
</tbody>
</table>

$$T_s = V^H \{ R_s^* \otimes R^H (H_{eq}) g_D g_D^H R(H_{eq}) \} V$$

$$t_r = V^H (E_s[s[n] - t] \otimes R^H (H_{eq}) g_D)$$

$$T_{dk} = \delta_k E \{ (s^* \otimes I_{MK}) I_{MK}[L_D + L_0] \}$$

$$T_{cd} = \epsilon_d V^H E \{ (s^* \otimes H^H_{eq}) I_{MK}[L_D] \}$$

$$T_{se} = \epsilon_s V^H E \{ (s^* \otimes H^H_{eq}) I_{MK}[L_D] \}$$

$$t = E \{ || s[n] ||^2 \} + g_D^H (\sigma_T^D I_{N_0(L_0+1)}) + \sigma_T^D \tilde{R}(H_{eq}) R(H_{eq}) R(H_{eq}) I_{N_0}[L_{RD}] (s^T \otimes H_{SR}) \} V$$

MSE as a function of $f_3 = \text{vec} \{ [g_5[0] \ldots g_5[L_3]] \}$


