A design framework for all-digital mmWave massive MIMO with per-antenna nonlinearities

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Abstract—Millimeter wave MIMO combines the benefits of compact antenna arrays with a large number of elements and massive bandwidths, so that fully digital beamforming has the potential of supporting a large number of simultaneous users with per user data rates of multiple gigabits/sec (Gbps). In this paper, we develop an analytical model for the impact of nonlinearities in such a system, and illustrate its utility in providing hardware design guidelines regarding two key challenges: the low available precision of analog-to-digital conversion at high sampling rates, and nonlinearities in ultra-high speed radio frequency (RF) and baseband circuits. We consider linear minimum mean square error (LMMSE) reception for a multiuser MIMO uplink, and provide performance guarantees based on two key concepts: (a) summarization of the impact of per-antenna nonlinearities via a quantity that we term the “intrinsic SNR”, (b) using linear MMSE performance in an ideal system without nonlinearities to bound that in our non-ideal system. For our numerical results, we employ nominal parameters corresponding to outdoor picocells operating at a carrier frequency of 140 GHz, with a data rate of 10 Gbps per user.

Keywords—All-digital massive MIMO uplink design, LoS channel, Nonlinearity (P_{non}), Low-precision ADC, Load factor, LMMSE.

I. INTRODUCTION

The potential of massive scale multiuser MIMO for meeting the ever-increasing demand for wireless mobile data is well understood [1], [2]. Massive MIMO becomes particularly attractive as we move up in the frequency spectrum toward millimeter wave (mmWave) and terahertz (THz) frequencies, where bandwidth is plentiful, and the wavelength is small enough to fit a large number of antennas on moderately sized platforms. By utilizing these advantages, mmWave massive MIMO can potentially support tens or hundreds of simultaneous users with per-user data rates of multiple gigabits/second (Gbps).

Two key bottlenecks to realizing this potential are the cost and power consumption of radio frequency (RF) frontends at high carrier frequencies and analog-to-digital conversion at large bandwidths. Due to the higher power consumption of high-speed RF chains, mmWave prototypes have thus far opted against fully digital arrays [3], and much of the recent research and development has focused on analog RF beamforming [4]–[9], supporting a single user at a time, or hybrid beamforming [10]–[15], where the number of supported users equals the number of RF chains, typically set to be much smaller than the number of antennas. However, recent advances in mmWave silicon hardware imply that scaling the number of RF chains with the number of antennas is on the cusp of feasibility, opening up the possibility of digital beamforming, where the number of supported users scales linearly with the number of antennas. By reducing dynamic range requirements and increasing amplifier loading, we can boost the power efficiency of each RF chain enough to allow scaling to fully digital arrays with hundreds of elements, but at the cost of increasing nonlinearity in the RF chain. Similarly, drastically reduced precision can be used to control the cost and power consumption of analog-to-digital conversion, as well as that of communication and computation on the digital backend. Robust system design in the presence of such nonlinearities therefore plays a critical role in scaling digital beamforming to mmWave massive MIMO. Our goal in this paper is to provide an analytical framework for quantifying the system-level impact of such nonlinearities, and to demonstrate how the increased degrees of freedom help relax linearity requirements and hardware specifications.

We consider, as a running example, a 256-element linear array at 140 GHz carrier frequency, with a symbol rate of 5 Gbaud with QPSK modulation for each supported user. For a load factor (defined as the ratio of the number of simultaneous users to the number of antennas) ranging from \( \frac{1}{16} \) (16 users) to \( \frac{1}{2} \) (128 users), the aggregate data rate ranges from 160 Gbps to 1.28 Tbps! However, scaling to this regime is challenging: wideband RF and baseband circuits scaled via relatively low-end silicon semiconductor processes (e.g., CMOS) exhibit significant nonlinearities, while the analog-to-digital converters (ADCs) available at multi-GHz sampling rates have relatively low precision. We provide in this paper an analytical framework that enables designers to determine the permissible levels of such nonlinearities for their desired system-level performance guarantees. As we shall show, the RF linearity and ADC precision requirements for load factor \( \frac{1}{2} \) are quite stringent, while reducing the load factor to \( \frac{1}{16} \) results in significantly relaxed hardware specifications. For a given number of users to be supported, therefore, a massive MIMO architecture can be leveraged to overcome severe nonlinearities, by increasing the number of antenna elements in order to reduce the load factor. We note that, besides the enormous aggregate throughput from supporting multiple simultaneous users, recent work also indicates that an all-digital solution can be more efficient in terms of hardware power consumption and...
area compared to a hybrid architecture [16].

A. Contributions

We provide design guidelines based on linear MMSE reception, with an analytical framework based on two core concepts: (a) We use a matched filter bound to show that the impact of per-antenna nonlinearities is effectively summarized by a quantity that we term the intrinsic SNR, corresponding to a normalized version of the nonlinearity. Key elements of this characterization are the well-known Bussgang decomposition, an overview of which can be found in [17], and the observation that, even for a moderate number of simultaneous users and without rich scattering, the antenna input is well modeled as zero-mean complex Gaussian. We show that the matched filter bound on the effective SNR for a given user, which captures the effect of the self-noise generated by per-antenna nonlinearities, depends only on four parameters: the user’s SNR, the intrinsic SNR, the load factor and a power control factor which summarizes the variations in received signal power across users. (b) We show that a pessimistic estimate of the degradation in performance due to multiuser interference can be obtained by analyzing (theoretically and/or numerically) an ideal system without nonlinearities. This enables us to provide a lower bound on the output signal-to-interference-plus-noise ratio (SINR) of a linear MMSE receiver, accounting for both nonlinearities and multiuser interference.

Combining these two concepts, averaging over the spatial distribution of users, and specializing to an edge user in the cell, allows us to provide analytical guidelines for maximum permissible levels of nonlinearities in order to provide the desired system-level performance guarantees (e.g., on outage probabilities). Consequently, our analysis utilizes the analytical capabilities of the Bussgang decomposition and LMMSE framework to provide a cross-layer design tool that links hardware level specifications to the system level performance metrics of a multi-user massive MIMO system. This enables exploration of fundamental tradeoffs between power consumption and cost of RF frontends and system performance. We consider third order RF and baseband nonlinearities that can be specified using the so-called 1 dB compression point [18], termed $P_{1\text{dB}}$. The per-antenna ADCs for the in-phase and quadrature components are modeled as overloaded uniform quantizers optimized (for a specified number of bits) to minimize the mean square error with zero mean Gaussian input. Using our framework, we are able to provide compact design prescriptions for $P_{1\text{dB}}$ and the number of ADC bits. For example, for a load factor of $1/2$, the system can work with 4-bit ADC and passband/baseband $P_{1\text{dB}}$ of 8.4 dB / 5 dB. On the other hand, 2-bit ADC with passband/baseband $P_{1\text{dB}}$ of 1.4 dB / -1 dB suffice to work properly with a load factor of 1/16. We present extensive simulations verifying our analytical predictions and prescriptions.

B. Related Work

While the focus in the present paper is on mmWave massive MIMO, there is a significant body of closely related recent research on the effect of nonlinearities on multiuser massive MIMO at lower carrier frequencies. Most of this prior work also employs Bussgang’s theorem [19] to model the effect of nonlinearities, both for uplink reception and downlink precoding. Our discussion here is limited to the literature on uplink massive MIMO, since that is the focus of the present paper, but the design framework for modeling downlink nonlinearities such as power amplifiers and digital-to-analog converters (DACs) is well known to be entirely analogous.

The line of sight (LoS) channel model used in our performance evaluation is different from that in much of this prior work, which employs models that are better matched to the propagation environments at lower frequencies. However, our analytical framework is quite general, and can be used to obtain design prescriptions for lower carrier frequencies as well. Conversely, many of the general observations emerging from prior work at lower carrier frequencies are consistent with the conclusions in the present paper, given a common underlying mathematical framework that employs the Bussgang decomposition and exploits the relaxation of hardware constraints enabled by the increase in the number of antennas. In the following, we briefly review this prior work in order to place the contributions of the present paper in perspective.

The potential for relaxing hardware constraints by increasing the number of antennas is clearly brought out by the theoretical results in [20], which show that the performance degradation due to hardware impairments vanishes asymptotically as the number of base station antennas gets large. The same trend holds for a finite but large number of antennas, as is clear from the results in [21]–[23], which study the spectral efficiency of quantized massive MIMO over frequency nonselective Rayleigh and Rician fading channels using maximum ratio combining. Another interesting conclusion from the simulations of [22] is that, for Rician fading, the system is more vulnerable to drastic quantization as the relative strength of the dominant component increases. Thus, the LoS model considered in this paper may be a worst-case scenario for obtaining design prescriptions regarding nonlinearities.

The impact of imperfect power control for quantized massive MIMO over frequency nonselective channels is included in the analysis in [24], [25]. Using spectral efficiency as a performance measure, an example conclusion from [24] is that 3-bit ADC suffices for a system with 100 antennas serving 10 users at a spectral efficiency of 3.5 bits per channel use, with 4-bit ADCs recommended to handle imperfections in power control and automatic gain control. Similar conclusions are obtained in [25], which shows moderate drops in spectral efficiency due to imperfect power control.

The impact of quantization on multiuser OFDM MIMO over a frequency-selective channel is studied in [26], with a focus on low-complexity channel estimation and data detection. The simulations in this paper show that, for the models considered, 4-bit ADC is sufficient to achieve a near-optimal performance (in terms of packet error rate) for a load factor of $1/8$ or lower. More recent work with a similar model [27] employs a Bussgang-based analysis for the joint distortion introduced by nonlinear low-noise amplifiers, phase noise, and finite-resolution ADCs, and demonstrates its accuracy by comparing
analytical predictions with simulations.

In comparison with the existing literature, the key conceptual novelty in the present paper is that we provide an analytical framework for mapping system-level performance goals to hardware design prescriptions for per-antenna nonlinearities. The theoretical foundation for this mapping is our observation (Theorem V.2) that an ideal system without linearities. The theoretical foundation for this mapping is

\[
\mathbf{H} = [\mathbf{h}_1 \mathbf{h}_2 \ldots \mathbf{h}_K]
\]

where \( \mathbf{H} \) is the channel matrix, \( \mathbf{x} = [x_1, \ldots, x_K]^T \) is the vector of symbols (normalized to unit energy: \( E[|x_k|^2] = 1 \)) transmitted by the mobiles, \( \mathbf{n} \sim \mathcal{CN}(0, \sigma_n^2 \mathbf{I}) \) is the thermal AWGN vector, and \( g(\cdot) \) is the effective per-antenna nonlinearity in complex baseband.

Running example: As mentioned in the introduction, for our running example, we set \( N = 256 \), with load factor \( \beta \) ranging from \( \frac{1}{16} \) to \( \frac{1}{2} \) (i.e., \( K \) ranging from 16 to 128). We provide the link budget analysis for the envisioned system in Appendix A to highlight the feasibility of low-cost silicon hardware realizations.

In the remainder of this section, we characterize the statistics of the received signal at each antenna and describe the nonlinearity models included in our numerical results.

A. Per-antenna Received Signal Statistics

The input to the effective complex baseband nonlinearity \( g(\cdot) \) at, say, antenna \( m \), is given by

\[
y_m = \sum_{k=1}^{K} A_k e^{j\phi_k} x_k e^{jm\Omega_k}.
\]

For a uniform spatial distribution of users over the region of interest, the amplitudes \( \{A_k\} \) and spatial frequencies \( \{\Omega_k\} \) are independent and identically distributed (i.i.d.). The phases \( \{\phi_k\} \) are uniform over \([0, 2\pi]\), and \( x_k \) are i.i.d. QPSK symbols. By virtue of the central limit theorem (CLT), the received signal is well modeled as zero-mean complex Gaussian for large \( K \), and jointly Gaussian across antennas. We have verified empirically, via histogram comparisons, quantile-quantile plots and KL divergence computations, that this Gaussian approximation holds for even moderate number of mobiles (e.g., \( K = 8 \)) in all settings that we have considered. Fig. 2 (a) illustrates a comparison between the histogram of the normalized real/imaginary component of the received signal and the standard normal distribution \( \mathcal{N}(0, 1) \).

In terms of technical conditions for applying the CLT, we note that it holds for independent, non-identically distributed

![Diagram](attachment:image.png)

**Figure 1:** The cell size is constrained radially between \( R_{\text{min}} \) and \( R_{\text{max}} \) and angularly between \(-\pi/3 \leq \theta \leq \pi/3\). BW_{3dB} stands for the 3 dB beamwidth. The passband and baseband nonlinearities are modeled by saturated third order polynomials. An overloaded uniform ADC with \( b \) bits per dimension, optimized for a zero-mean standard Gaussian random variable, is used. Linear MMSE reception is employed after digitization.

II. System Model

Fig. 1 shows the system model. The base station performs horizontal scanning with a 1D half-wavelength spaced \( N \)-element array. Let \( K \) denotes the number of simultaneous users, and \( \beta = \frac{K}{N} \) the load factor.

We assume a line-of-sight (LoS) channel between the base station and each mobile. The direction of arrival (DoA) from the \( k \)-th mobile is denoted by \( \theta_k \), and corresponds to spatial frequency \( \Omega_k = 2\pi d_e \sin \theta_k \), where \( \lambda \) denotes the carrier wavelength and \( d_e \) denotes the inter-element spacing, set to \( \frac{\lambda}{2} \) in our numerical results. The \( N \times 1 \) spatial channel for mobile \( k \) is given by

\[
\mathbf{h}_k = A_k e^{j\phi_k} [1 e^{j2\Omega_k} \ldots e^{j(N-1)\Omega_k}]^T,
\]

where \( \phi_k \) is an arbitrary phase shift and \( A_k^2 = \left( \frac{\lambda}{2 \pi R_k} \right)^2 \) depends on the radial location \( R_k \) of mobile \( k \), using the Friis formula for path loss.

The cascade of the nonlinearities described in Sections II-B and II-C is modeled as a complex baseband equivalent nonlinearity \( g(\cdot) \). The complex baseband received signal vector \( z \) at the base station is therefore given by

\[
z = g(y) = g \left( \frac{\mathbf{H}}{\sqrt{N \times K}} \mathbf{x} + \mathbf{n} \right),
\]
The 1 dB compression point \( P_{\text{1dB}} \) is defined as the input power at which the output power of the desired sinusoid (at \( f_o \)) is compressed by 1 dB.

Figure 3: The 1 dB compression point \( P_{\text{1dB}} \) is defined as the input power at which the output power of the desired sinusoid (at \( f_o \)) is compressed by 1 dB.

Random variables if the variances are bounded [33], which is the case here: with power control, the amplitudes of \( \{A_k\} \) are tightly clustered, whereas with no power control they lie within a range of values determined by the maximum and minimum link distance. The randomness in the terms of the sum in (3) results from randomness in channel phases and data modulations, and CLT is applied conditioned on \( \{A_k\} \). We average over the realizations of \( \{A_k\} \) to determine expected receiver performance and reliability bounds.

B. Passband and Baseband Nonlinearity Model

The passband nonlinearity arises in the low noise amplifier and the mixer, while the baseband nonlinearity is in the variable gain amplifier. We model each nonlinearity as a saturated third-order polynomial function with a nominal gain of unity. The function is parametrized by the 1 dB compression point \( P_{\text{1dB}} \) [18], defined as the input power of a sinusoid of frequency \( f_o \) (taken to be the carrier frequency) at which the output power is reduced by 1 dB relative to the nominal. The concept is illustrated in Fig. 3. A commonly used model for gain saturation using third-order nonlinearity is the cubic soft clipper, which can be parametrized by \( P_{\text{1dB}} \) as follows:

\[
g(y(t)) = \begin{cases} 
  y(t) - \frac{0.44 |y(t)|^2}{P_{\text{1dB}}} & \text{if } |y(t)|^2 \leq \frac{P_{\text{1dB}}}{0.44} \\
  y(t) | \frac{y(t)}{|y(t)|} | \sqrt{P_{\text{1dB}}} & \text{if } |y(t)|^2 > \frac{P_{\text{1dB}}}{0.44}
\end{cases} \tag{4}
\]

The gain compression for the passband nonlinearity depends on the absolute value of the complex baseband signal, while the gain compression depends on the absolute value of the I and Q components for the baseband nonlinearity. Fig. 4 (a) illustrates the distribution of the input powers of the passband and baseband nonlinearities, along with example input/output (I/O) characteristics. In this work, we consider the nonlinearities to be memoryless and free of phase distortion.

C. ADC Model

After down-conversion, each baseband component is quantized to \( b \) bits by an ADC. We design the quantizer to minimize the mean squared error (MSE) assuming that the incoming signal is Gaussian with zero mean and unit variance. An automatic gain control (AGC) precedes the ADC in order to normalize the average power of the input signal to unity, and ensure that the full dynamic range of the ADC is utilized. We employ an overloaded uniform ADC [34]: while the MSE could be improved slightly by designing a non-uniform quantizer for standard Gaussian input, the improvement is slight and has no discernible impact on system-level performance (see Appendix B for a quantitative discussion). Fig. 4 (b) depicts a 4-bit uniform overloaded quantizer.

D. Linear MMSE Detector

We show in a following section that the impact of a per-antenna nonlinearity \( g(\cdot) \) can be modeled as additional noise,
leading to an equivalent system model of the form
\[ y = Hx + \tilde{u}, \] (5)
where \( \tilde{u} \) is zero mean with variance \((\sigma_n^2 + \nu_g^2)I\). The value of \( \nu_g \) is specified in section V. Thus, any adaptive implementation of the linear MMSE receiver automatically accounts for the nonlinearities. The linear MMSE receiver is specified as follows:
\[ \hat{x} = Wy, \] (6)
where
\[ W = (H^H H + (\sigma_n^2 + \nu_g^2)I)^{-1} H^H. \] (7)

The linear MMSE detector has a rich history with well-known exposition, see [35], [36]. In order to provide a self-contained exposition, we review the Bussgang linearization in the context of our MIMO system.

III. Bussgang Linearization

In order to provide a self-contained exposition, we review Bussgang linearization in the context of our MIMO system.

A. Scalar Bussgang Linearization

For a zero mean complex-valued random variable \( y \) and a nonlinearity \( g(\cdot) \), a linear MMSE approximation of \( g(y) \) by \( ay \) satisfies the orthogonality principle [37]:
\[ \mathbb{E}((g(y) - ay)y^*) = 0. \] (8)
Standard computations for the linear gain \( a \) and the variance of the approximation error \( e = g(y) - ay \) yield
\[ a = \frac{\mathbb{E}(g(y)y^*)}{\mathbb{E}(|y|^2)}, \] (9)
\[ \sigma_g^2 = \mathbb{E}(|e|^2) = \mathbb{E}((g(y))^2) - |a|^2\mathbb{E}(|y|^2). \] (10)
Hence, \( g(y) \) can be written as
\[ g(y) = ay + e, \] (11)
where \( a \) and \( \mathbb{E}(|e|^2) = \sigma_g^2 \) can be computed analytically or empirically for any distribution of \( y \) and nonlinear function \( g(\cdot) \). Bussgang evaluated \( a \) and \( \sigma_g^2 \) for different nonlinear functions when the input \( y \) is Gaussian random variable [19].

B. Vector Bussgang Linearization

The main part of Bussgang’s theorem in [19], and its extension to the complex domain in [38], is the preservation of covariance structure under nonlinearities for jointly Gaussian random variables: If \( y \) and \( z \) are jointly Gaussian random variables and \( g(\cdot) \) is a nonlinear function, then \( \mathbb{E}(g(y)z^*) = a\mathbb{E}(yz^*) \), where \( a \) is defined in (9).

This result allows us to characterize the linear MMSE fit for a Gaussian random vector in terms of the scalar linear MMSE fits for its components. It has been customized to MIMO in many recent papers [25], [27], [39], [40], hence we state the relevant result here without proof (see Appendix A in [40] for a derivation).

Theorem III.1. Vector Bussgang Decomposition

Let \( y \) denote the jointly Gaussian random vector input to the effective nonlinearity \( g(\cdot) \) referred to complex baseband, so that the received signal \( z = g(y) \). Then the Bussgang decomposition of \( z \) is given by
\[ z = g(y) = Ay + e, \] (12)
where
\[ A = \text{diag}([a_1, \ldots, a_N]), \] (13)
\[ a_i = \frac{\mathbb{E}(g(y_i)y_i^*)}{\mathbb{E}(|y_i|^2)}, \] (14)
and the variance of element \( e_i \) of the approximation error vector \( e \) is given by
\[ \sigma_{g_1}^2 = \mathbb{E}(|g(y_i)|^2) - |a_i|^2\mathbb{E}(|y_i|^2). \] (15)

The Bussgang theorem on covariance preservation therefore leads to a linear MMSE fit with diagonal covariance structure. Moreover, the diagonal elements are equal if the statistics of \( \{y_i\} \) are identical, as in the following straightforward corollary, stated without proof.

Corollary 1. If the diagonal elements of the covariance of \( y \) are equal, i.e., \( \mathbb{E}(|y_i|^2) = \mathbb{E}(|y_k|^2), \forall i, k \), then the Bussgang decomposition specializes to
\[ z = g(y) = ay + e, \] (16)
where \( a \) and \( \mathbb{E}(|e|^2) = \sigma_g^2 \) are the scalar Bussgang parameters of \( g(\cdot) \).

It is worth noting that the self-noise \( e \) may be spatially correlated. However, recent work [39] indicates that this correlation becomes negligible when the number of users is large, and we ignore it in our analysis here.

IV. Bussgang Normalization and Intrinsic SNR

In this section, we define a normalization such that the Bussgang parameters for a nonlinearity are independent of input power. We introduce the concept of intrinsic SNR to characterize the self-noise in this normalized setting. As we shall see, this is the summary specification that is provided by system-level design requirements to the hardware designer,
Figure 5: The nonlinear function $g(\cdot)$ in (a) can be decomposed to the linear model in (b) whose parameters depend on the input power. We define a normalized version of the nonlinearity in (c), which allows us to provide design specifications independent of input power. The corresponding normalized linearization is depicted in (d).

Figure 6: (a) The conventional limiter function. (b) a unity-gain limiter function whose clipping threshold is normalized to the effective input power.

Figure 7: (a) Bussgang parameters and (b) the intrinsic SNR of the normalized limiter function.

As a simple example, consider a memoryless limiter as depicted in Fig. 6 (a), which is specified by the gain $G$ and the power threshold $P_{th}$ at which the output signal is clipped. The normalized version of this function has unity gain, as shown in Fig. 6 (b), hence we only need to specify a single parameter to characterize it: the clipping threshold $\tilde{P}_{th} = P_{th}/G^2\sigma^2_y$ normalized to the input power $\sigma^2_y$. The Bussgang parameters of the normalized limiter function are shown in Fig. 7 (a), and the intrinsic SNR is shown in Fig. 7 (b).

Henceforth, nonlinearities and their Bussgang parameters are normalized to the input power, and we drop the “tilde” notation to denote the normalized version. For example, the 1 dB compression point of a passband/baseband nonlinearity is normalized to the input power, and hence is measured in dB instead of dBm.

**Design Approach**

The analytical framework described in the next section leads the following design approach for going from system-level performance metrics to hardware design specifications:

- MIMO performance specifications lead to a requirement for the intrinsic SNR for the per-antenna nonlinearities, ignoring the specific nature of the nonlinearities. For example, suppose that we require an intrinsic SNR of 20 dB at least 95% of the...
time.

- We map the intrinsic SNR requirement to a specification for the normalized nonlinearity. Taking the limiter in Fig. 6 as an example, we see from Fig. 7 (b), the clipping threshold normalized to the effective input power, $P_{th}/G^2 \sigma_y^2$, must be at least 6 dB in order to attain an intrinsic SNR of 20 dB.
- In this step, the absolute value of the gain and clipping threshold is calculated. For example, suppose that the system in our running example is at load factor $\beta = 1/4$, i.e., 64 users. Then, according to the link budget presented in Appendix A, the input power to the receive chain is $-60$ dBm if power control is employed. We therefore obtain that $P_{th}/G^2 = -54$ dBm. The hardware designer now has to choose $G$ and $P_{th}$ in order to achieve this ratio or better.

V. ANALYTICAL FRAMEWORK

Our analytical framework is developed as follows.

1) We derive a matched filter bound for each user in the MIMO system that accounts for the self-noise due to the per-antenna nonlinearities (which scales with the power summed across users) as well as thermal noise. To this end, we use Bussgang linearization and the intrinsic SNR discussed in the previous section.

2) We derive a lower bound for the output SINR of the LMMSE receiver for any given user. Defining the efficiency of the LMMSE receiver for a given user as the ratio of SINR to SNR, we show that an efficient user in an ideal system without nonlinearities is a lower bound on that of the actual system. This, together with the matched filter bound, provides a lower bound on the LMMSE output SINR.

3) We obtain system-level design prescriptions by specializing the preceding lower bound to an “edge" user whose performance is stochastically poorer than that of any other user.

A. Bussgang Linearized Model

As described in section II, we denote by $\{A_k, k = 1, \ldots, K\}$ the amplitudes of the incoming waves for the $K$ users, and by $\sigma_n^2$ the variance of the thermal noise at each antenna. We can therefore model the incoming signal at each receive antenna as $y_m \sim CN(0, \sigma_y^2)$, where

$$\sigma_y^2 = \sum_{k=1}^K A_k^2 + \sigma_n^2 = \sigma_n^2 + KA_{rms}^2,$$  \hspace{1cm} (18)

and

$$A_{rms} = \sqrt{\frac{1}{K} \sum_{k=1}^K A_k^2}$$  \hspace{1cm} (19)

is the root mean square (rms) amplitude, averaged across users.

As depicted in Fig. 5 (c), using the normalized Bussgang linearization requires scaling the incoming signal to unit variance as follows:

$$\tilde{y}_m = y_m/\sigma_y.$$  \hspace{1cm} (20)

For a normalized nonlinearity $g(\cdot)$ as defined in the previous section, our per antenna linearized model is given by:

$$g(\tilde{y}_m) = a \tilde{y}_m + e_m.$$  \hspace{1cm} (21)

For the received signal (2), the normalized signal prior to passing through the nonlinearity is given by

$$\tilde{y} = y / \sigma_y.$$  \hspace{1cm} (22)

Using the Bussgang decomposition, we have

$$g(\tilde{y}) = a \tilde{y} + e = a y + e,$$  \hspace{1cm} (23)

where $E[ee^H] = \sigma_e^2 I$. We can now go back to the original signal scaling to obtain

$$\tilde{y} = \frac{\sigma_y}{a} g(\tilde{y}) = y + \frac{\sigma_y}{a} e = Hx + n + \frac{\sigma_y}{a} e.$$  \hspace{1cm} (24)

This is the model (5), with effective noise

$$\tilde{n} = n + \frac{\sigma_y}{a} e$$  \hspace{1cm} (25)

of variance

$$E(\tilde{n} \tilde{n}^H) = (\sigma_n^2 + \nu^2) I$$  \hspace{1cm} (26)

where

$$\nu^2 = \frac{\sigma_y^2}{|a|^2} \frac{\sigma_y^2}{\gamma_g}.$$  \hspace{1cm} (27)

B. Matched Filter Bound

For the $k^{th}$ user, the matched filter bound for the linearized model (5), with equivalent noise as in (24)-(25), is simply given by

$$\text{SNR}_k(g) = \frac{||h_k||^2}{\sigma_n^2 + \nu^2}.$$  \hspace{1cm} (28)

Our design framework is built around the dependence of this bound on key system parameters as stated in the following theorem. We first ignore thermal noise, in order to clearly bring out the role of intrinsic SNR, $\gamma_g$, and load factor, $\beta$, and then include its effect.

Theorem V.1. Matched filter bound

(a) Self-noise only: Ignoring thermal noise, the matched filter bound for user $k$ is given by

$$\text{SNR}_k(g) = \frac{\gamma_g A_k^2}{\beta A_{ms}^2}.$$  \hspace{1cm} (29)

(b) Self-noise and thermal noise: The matched filter bound for user $k$, considering both self-noise and thermal noise, is given by

$$\text{SNR}_k(g, \sigma_n^2) = \frac{1}{\text{SNR}_k(g) + \frac{1}{\gamma_g} + \frac{1}{\text{SNR}_k}}.$$  \hspace{1cm} (30)

where $\text{SNR}_k = N A_k^2 / \sigma_n^2$ is the SNR for user $k$ accounting for thermal noise alone.
Proof: The proof involves algebraic manipulations based on the linearized model (24)-(25).
(a) Using (1), the numerator in (28) is given by
\[ |h_k|^2 = NA_k^2. \]  
Using (18) and (27), and setting \( \sigma_n^2 = 0 \), the denominator in (28) is given by
\[ \nu^2_g = \frac{K A_{\text{rms}}^2}{\gamma_g}. \]  
Plugging (31) and (32) into (28), we obtain
\[ \text{SNR}_k(g) = \frac{NA_k^2 \gamma_g}{KA_{\text{rms}}^2} = \frac{\gamma_g A_k^2}{\beta A_{\text{rms}}^2}, \]  
which is the desired result (29).
(b) From (28) and (31), we have
\[ \frac{1}{\text{SNR}_k(g, \sigma_n^2)} = \frac{\sigma_n^2}{NA_k^2} + \nu^2_g. \]  
For non-zero thermal noise, we have, using (18) and (27), that
\[ \nu^2_g = \frac{KA_{\text{rms}}^2 + \sigma_n^2}{\gamma_g}. \]  
Plugging into (34), we obtain upon simplification the desired result (30).

Note that, if \( \gamma_g \gg 1 \), then the formula (30) reduces to
\[ \text{SNR}_k(g, \sigma_n^2) = \frac{1}{\text{SNR}_k(g) + \text{SNR}_k}. \]  

In order to provide system-level performance guarantees, we focus on supporting users at the cell edge. We therefore set \( A_k \) to the worst-case amplitude \( A_{\text{edge}} \) (at 100 m range for our running example), while computing \( A_{\text{rms}} \) by a statistical average \( \sqrt{\mathbb{E}[A_k^2]} \) given the users distribution, assuming a large enough number of users. Let us term the ratio of the power of the edge user to the rms power as the power control factor, since it depends on the power control scheme used. The power control factor \( \alpha_p \) is given by
\[ \alpha_p = \frac{A_{\text{edge}}^2}{A_{\text{rms}}^2}. \]  
Specializing (29) to the edge user, we now obtain that
\[ \text{SNR}_{\text{edge}}(g) = \gamma_g \frac{1}{\beta} \alpha_p. \]  

Power control factor with no power control: For users who are uniformly distributed over the area bounded by \([R_{\text{min}}, R_{\text{max}}]\) and a given angular range, we obtain upon straightforward computation that, for a system without power control,
\[ \alpha_p = \frac{1}{\log \frac{R_{\text{min}}}{R_{\text{max}}}} \int_{R_{\text{min}}}^{R_{\text{max}}} \frac{1}{r^4} dr, \]  
\[ = \frac{1 - R_{\text{min}}^2}{2 \log \frac{R_{\text{max}}}{R_{\text{min}}}}. \]  
which evaluates to -7.8 dB for \( R_{\text{max}} = 100 \) m, \( R_{\text{min}} = 5 \) m.

C. Lower Bound on LMMSE Output SINR

We now provide a lower bound on the output SINR of any user via the ideal system.

**Theorem V.2. LMMSE Lower Bound**

In the presence of nonlinearity, a lower bound on the output SINR of a linear MMSE receiver for any user is given as

\[ \text{SINR} \geq \text{SNR}(g, \sigma_n^2) \eta_{\text{ideal}}, \]  
where
\[ \eta_{\text{ideal}} = \frac{\text{SNR}(\text{ideal})}{\text{SNR}(\text{ideal})}. \]

is the efficiency in an ideal system with the same user configuration and amplitudes, but without nonlinearity.

Proof: Since the system described in (5) is a pessimistic model for the system in (2), we have by Lemma C.1 in Appendix C that

\[ \frac{\text{SINR}}{\text{SNR}(g, \sigma_n^2)} \geq \frac{\text{SNR}(\text{ideal})}{\text{SNR}(\text{ideal})}, \]  
where \( \text{SNR}(\text{ideal}) \) is the target linear MMSE output SINR for a user in an ideal system (without nonlinearities).

We evaluate \( \eta_{\text{ideal}} \) through simulations of the ideal system for the edge user, as shown in Fig. 8, where \( \text{SNR}_{\text{edge}} = \frac{N A_{\text{edge}}^2}{\sigma_n^2} \), and \( A_{\text{edge}} \) is the received amplitude of the user at 100 m. The target output SINR of the linear MMSE, i.e., \( \text{SNR}_{\text{edge}}(\text{ideal}) \) is 9.7 dB. This number corresponds to the \( \text{SNR}_{\text{edge}} \) in a single user case. Hence, in a single user case \( \eta_{\text{ideal}} = 1 \). As the load factor increases, there is noise enhancement due to interference suppression: \( \eta_{\text{ideal}} = 9.7 - \text{SNR}_{\text{edge}} \) can be inferred from Fig. 8 (b).

D. From System-Level Performance to Intrinsic SNR

The chosen quality of service measure maps to an SINR requirement at the LMMSE output. We compute this for the
ideal system. For example, simulating the ideal system, a target BER of $10^{-3}$ with 95% availability is obtained for $\text{SNR}_{\text{edge}}(\text{ideal}) = 9.7$ dB. Since the SNR for an edge user is 14 dB, we see from Fig. 8 (b) that the efficiency for the ideal system is given by $9.7 - 14 = -4.3$ dB for no power control and $\beta = 1/2$. This is an upper bound on the efficiency of the actual system.

We can now compute the minimum SNR $\text{SNR}_{\text{edge}}(g, \sigma_n^2)$ from Eq. (40) to achieve the required SNR in the presence of nonlinearities. Finally, we can infer the intrinsic SNR $\gamma_g$ required from Eq. (30) and Eq. (36). This is now mapped to detailed hardware specifications, as illustrated by examples in the next section.

VI. DESIGN EXAMPLES AND PERFORMANCE EVALUATION

The system parameters are as described in Section II. We illustrate our design for a target uncoded BER of $10^{-3}$, which is low enough for reliable performance using a high-rate channel code with relatively low decoding complexity. For QPSK, the corresponding required SNR over a SISO AWGN link is 9.7 dB. This becomes our target SNR at the output of the LMMSE receiver for an edge user. This setting is simply for illustration: our analytical framework applies for any QoS measure that can be approximated in terms of SNR (e.g., outage capacity or spectral efficiency using Shannon’s formula).

In the following, we first describe the user distribution and power control schemes deployed in the cell. Then, we apply the analytical design framework to define the specification on the receive chain: the passband/baseband nonlinearity and the ADC resolution. We then evaluate the efficacy of the framework in attaining the desired system-level performance by simulations for selected scenarios. Finally, we provide design guidelines on the receive chain requirements in a more comprehensive set of scenarios.

A. User Distribution

The mobiles are uniformly distributed inside a region bordered by a minimum and a maximum distance away from the base station, $R_{\text{min}}$ and $R_{\text{max}}$, respectively. Since $\frac{d\Omega}{d\theta} \sim \cos \theta$, the spatial frequency is less responsive to changes in DoA for $\theta$ near $\pm \pi/2$, which makes it more difficult to separate mobiles towards the edge of the angular field of view. We therefore confine the field of view for the antenna array to $-\pi/3 \leq \theta \leq \pi/3$. While the mobiles are placed randomly in our simulations, we enforce a minimum separation in spatial frequency between any two mobiles in order not to incur excessive interference, choosing it as half the 3 dB beamwidth: $\Delta\Omega_{\text{min}} = \frac{2.783}{N}$ [41] (mobiles closer in spatial frequency could be served in different time slots, for example). An example distribution of mobiles is depicted in Fig. 9.

B. Power Control Schemes

Our analysis in Section V first considers a system with no power control, in which each transmitter transmits at equal power. We then consider two power control schemes: a naive scheme in which transmitters adjust their powers to be roughly equal at the receiver, to within a tolerance, and an adaptive power control scheme designed for the linear MMSE receiver [42] aimed at meeting an SNR target for each mobile at the receiver. Power control is a very well-studied area, hence our goal is to provide quick insight on its implications for our system, rather than performing a comprehensive evaluation.

1) Naive power control: In this scheme, the base station asks all the users to decrease their power to make their received power at the base station equal the received power of the farthest mobile, i.e., at $R_{\text{max}}$. A disadvantage of this scheme, illustrated by our performance results in subsequent subsections, is that nearby users are no longer able to use their larger signal strength to overcome the impact of interference from other users who are nearby (in terms of spatial frequency separation). The power factor $\alpha_p$ of the naive power control scheme is equal to 0 dB because all the users have the same received signal strength.

2) Adaptive power control: In order to avoid the pitfalls of naive power control, we consider an adaptive power control scheme aimed at meeting an SNR target $\text{SNR}_{\text{th}}$ at the output of the linear MMSE receiver. This was proposed and shown to converge in [42]. We restate it with all values represented in the dB scale in Algorithm 1. Starting from no power control and all users transmitting at maximum power, the algorithm seeks to enforce a threshold SNR, termed $\text{SNR}_{\text{th}}$, iteratively as follows: every mobile with $\text{SNR} = \text{SNR}_{\text{th}}$ reduces its power by $\text{SNR}_{\text{th}}$; every mobile with $\text{SNR} > \text{SNR}_{\text{th}}$ reduces its power by $\text{SNR} - \text{SNR}_{\text{th}}$. This process is repeated until a convergence criterion is met. The power factor $\alpha_p$, that results from this adaptive power control scheme is found via simulation to equal about $-2$ dB for our running example.

C. Applying the Design Framework

For illustration, we consider $\beta = \frac{1}{2}$ and $\beta = \frac{1}{16}$, with no power control, naive power control and adaptive power control.
Algorithm 1: Adaptive power control

Input: H, \{P_k^{(1)}\}_{k=1,...,K}
parameter: SINR_{th}, \epsilon
Output: \{P_k\}_{k=1,...,K}
1 Margin = \epsilon + 1;
2 while Margin > \epsilon do
3 \{SINR_k\}_{1:K} \leftarrow calculate LMMSE output SINR;
4 for k \leftarrow 1 to K do
5 \Delta SINR_k \leftarrow \max\{SINR_k - SINR_{th}, 0\};
6 P_k \leftarrow P_k - \Delta SINR_k;
7 Margin \leftarrow \max\{\Delta SINR_k\}_{1:K};

The design steps are as follows:
1) System-level design: We require SINR_{edge}(ideal) \approx 10 dB for our target QoS. Using simulations for the ideal system, we compute the LMMSE efficiency \eta_{ideal} as shown in Fig. 8. For our four scenarios, the LMMSE efficiency \eta_{ideal} is found to be (a) 4.5 dB, (b) 0 dB, (c) 4.5 dB, and (d) 0 dB. After that, we determine the SNR of the edge mobile and the intrinsic SNR jointly to achieve the LMMSE lower bound. Specifically, the contours in Fig. 10 (a) illustrates the following equation for each scenario:

\[
\text{SNR}(g, \sigma^2_n) = \frac{\text{SNR}_{edge}}{\eta_{ideal}},
\]

\[
= \frac{1}{\beta g \sigma_p^2 + \frac{1+g}{\gamma_g} \frac{1}{\text{SNR}_{edge}}} \frac{1}{\eta_{ideal}}.
\]

We pick the following combinations of (SNR_{edge}, \gamma_g): (a) (20,20) dB, (b) (11,12) dB, (c) (16,17.5) dB, and (d) (12,7) dB.

2) Hardware-level design: This step determines the specifications of the passband/baseband nonlinearity and the ADC to achieve the required intrinsic SNR. Fig. 10 (b) shows the tradeoff between the number of ADC bits and the 1 dB compression point of the baseband nonlinearity P_{bb1dB} and the passband nonlinearity P_{pb1dB}. The 1 dB compression points computed are normalized to the input power. The absolute compression points in dBm are computed by determining the average received input power at each base station antenna.

Here we have taken the link budget, or attainable SNR_{edge}, as our constraint, and have designed the nonlinearity specifications accordingly. We can, of course, utilize the same framework to determine the link budget required for a given set of nonlinearities.

D. Simulation-based Verification

Here, we verify the designs produced by our analytical framework by numerical simulations. In Fig. 11, we plot the BER that 95% of the users attain for the cases we mention in the previous subsection. As shown, all the curves reach the \(10^{-3}\) at slightly smaller SNR_{edge} than predicted by our analytical framework, which shows that our approach is both conservative and accurate. Table I summarizes our design prescriptions for different scenarios. As shown in the table, we examine the combination of four load factors with no power control and two power control strategies. We demonstrate the specification of the receive chain along with the resultant intrinsic SNR, \gamma_g. Then we compute an upper bound for the SNR needed for the edge user to achieve the performance metric. Finally, using simulations, we show the accuracy of the derived upper bound.

The results reported in Table I show that massive MIMO is key to relaxed hardware specifications: increasing the number of antennas for a given number of users reduces load factor, providing a “degrees of freedom” advantage that is used to level out the distortions caused by nonlinearities. For example,
Table I: This table presents the analytical predictions and simulation results for the SNR budget needed to meet the desired performance criterion (10^{-3} BER at 95% availability) for different scenarios. The intrinsic SNR $\gamma_g$ corresponds to the cascade of the passband and baseband nonlinearities, specified by their 1 dB compression points ($P_{dB}^{bb}$ and $P_{dB}^{pb}$, respectively), together with 6-bit ADCs for I and Q. PC and $\beta$ denote the power control scheme used, and the load factor, respectively.

<table>
<thead>
<tr>
<th>PC</th>
<th>$\beta$</th>
<th>$b$</th>
<th>$P_{dB}^{bb}$ (dB)</th>
<th>$P_{dB}^{pb}$ (dB)</th>
<th>$\gamma_g$ (dB)</th>
<th>SNR$_{edge}$ (upper bound)</th>
<th>SNR$_{edge}$ (sum.)</th>
</tr>
</thead>
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<tr>
<td>none</td>
<td>1/2</td>
<td>5</td>
<td>8.4</td>
<td>6.7</td>
<td>29</td>
<td>20</td>
<td>17.5</td>
</tr>
<tr>
<td>naive</td>
<td>1/2</td>
<td>4</td>
<td>8.4</td>
<td>4.9</td>
<td>17.5</td>
<td>18.7</td>
<td>18.4</td>
</tr>
<tr>
<td>adaptive</td>
<td>1/2</td>
<td>4</td>
<td>8.4</td>
<td>4.9</td>
<td>17.5</td>
<td>16</td>
<td>14.7</td>
</tr>
<tr>
<td>none</td>
<td>1/4</td>
<td>3</td>
<td>3.7</td>
<td>1.9</td>
<td>11.5</td>
<td>12.5</td>
<td>11.9</td>
</tr>
<tr>
<td>adaptive</td>
<td>1/4</td>
<td>3</td>
<td>3.4</td>
<td>1.9</td>
<td>11.5</td>
<td>12.5</td>
<td>11.9</td>
</tr>
<tr>
<td>none</td>
<td>1/8</td>
<td>3</td>
<td>4.2</td>
<td>1.4</td>
<td>12</td>
<td>13</td>
<td>12.2</td>
</tr>
<tr>
<td>naive</td>
<td>1/8</td>
<td>3</td>
<td>2.2</td>
<td>-1.1</td>
<td>8.7</td>
<td>12.7</td>
<td>12.7</td>
</tr>
<tr>
<td>adaptive</td>
<td>1/8</td>
<td>3</td>
<td>3.2</td>
<td>1.9</td>
<td>11</td>
<td>10.9</td>
<td>10.8</td>
</tr>
<tr>
<td>none</td>
<td>1/16</td>
<td>3</td>
<td>4.2</td>
<td>1.4</td>
<td>12</td>
<td>11.2</td>
<td>10.8</td>
</tr>
<tr>
<td>naive</td>
<td>1/16</td>
<td>2</td>
<td>1.4</td>
<td>-1.1</td>
<td>7.6</td>
<td>11.5</td>
<td>11.5</td>
</tr>
<tr>
<td>adaptive</td>
<td>1/16</td>
<td>2</td>
<td>-1.1</td>
<td>-1.9</td>
<td>7</td>
<td>11.8</td>
<td>11.2</td>
</tr>
</tbody>
</table>

when serving 16 users with a 256 element array ($\beta = \frac{1}{16}$), only 2 bits of quantization per dimension is required and the 1-dB compression point of RF and baseband amplifiers is very low (both below 0 dB with adaptive power control). In practice, a lower compression point allows higher loading, i.e., the amplifier can support a larger input signal and produce a stronger signal in the output, which increases the power efficiency of an amplifier. This is very desirable trait for scaling to large arrays. The results also bring out the impact of power control. Higher disparities in user powers requires higher ADC granularity to allow effective interference suppression of strong users which might otherwise “drown out” weak users. The reduction in user power disparities and required dynamic range with better power control is reflected clearly in our results.

VII. CONCLUSION

The analytical framework provided in this paper is a conservative, yet accurate, approach for designing hardware specifications for nonlinear elements in all-digital mmWave massive MIMO. Scaling using a larger number of antennas with a smaller load factor is attractive, since the specifications for RF nonlinearities, baseband nonlinearities, and ADC precision can all be significantly relaxed by operating at lower load factors. The requirements can also be relaxed by use of appropriate power control, as illustrated by the simple adaptive power control scheme considered here.

While we have considered LoS channel models here, we note that our approach extends to sparse multipath channels. At high symbol rates, equalization over a large delay spread becomes computationally unattractive. In this case paths that differ significantly in delay and angular spread from the dominant path play the role of additional interference, and can be folded into our framework.

In addition to the extensive effort required to realize our design prescriptions in hardware, there are also important open issues related to the digital backend, given the challenges of both computation and data transport for the multiGigabaud, multiuser system considered here. Thus, despite the extensive prior research on multiuser detection, there are significant open issues on the design of strategies that are efficient enough (in terms of both computation and communication on the backend fabric) to scale with the number of antennas, number of users, and bandwidth. We also note the importance of exploring nonlinear reception techniques (such as interference cancellation) that could outperform LMMSE detection while maintaining low computational overhead. Preliminary results in [43]–[45] indicate that exploiting channel sparsity is a promising approach for developing such techniques.

APPENDIX A

ALL-DIGITAL LINK BUDGET

We provide here example parameters that demonstrate that the link budget for all-digital massive multiuser MIMO uplink system is realizable with low-cost silicon:

- antenna element gain covering a hemisphere is 3 dBi,
- 16-element array at the mobile gives 12 dB transmit beamforming gain, plus 12 dB power pooling gain,
- 256-element array in the base station gives 24 dB receive beamforming gain,
- noise figure for each RF chain in the base station of 7 dB,
- thermal noise power over 5 GHz bandwidth is about -77 dBm,
- and free space path loss of an edge user at 100 m using a carrier frequency of 140 GHz is about 115 dB.

The transmit power required from each power amplifier (PA) at the mobile to achieve a target SNR (in dB) for an edge-user, namely $\text{SNR}_{edge|\text{dB}}$, can now be computed as

$$P_{PA} = \text{SNR}_{edge|\text{dB}} - 9 \text{ dBm}. \quad (43)$$

For example, $\text{SNR}_{edge|\text{dB}}$ of about 16 dB (shown to suffice for our case study) requires 7 dBm PA output, which is realizable in CMOS (CMOS designs of up to 11 dBm have been reported in [46]).

APPENDIX B

UNIFORM VS NONUNIFORM QUANTIZATION

Our simulation results are for an overloaded ADC. The overloaded uniform ADC comprises two regions in its I/O characteristic, the granular and overload regions. The granular region is quantized uniformly, with bounded quantization noise. While quantization noise in the overload region, represented by the quantizer levels at the edges, is unbounded, the contribution to the MSE is kept comparable to that of the granular region by minimizing the MSE for the given input distribution. Note that the MSE is plotted against overload threshold.

An alternative is to employ an MSE-optimal quantizer using Lloyd’s algorithm [47], with quantization bins as listed in [48]. The MSE comparison between these two options is shown in Fig. 12 (b). The advantage of nonuniform MSE-optimal
quantization is barely noticeable for the small number of quantization bits of interest here, hence we choose to work with the simpler overloaded uniform quantizer.

APPENDIX C
LINEAR MMSE PROPERTIES

From the point of view of a given user (the desired user) with channel \( \mathbf{h} \), we may write the received signal corresponding to a single symbol as

\[
\mathbf{r} = \mathbf{s} \mathbf{h} + \mathbf{w}_I + \mathbf{w}_N,
\]

where \( \mathbf{s} \) denotes the transmitted symbol, \( \mathbf{w}_I \) denotes the interference vector and \( \mathbf{w}_N \) is the zero mean noise vector with covariance matrix \( \sigma_I^2 \mathbf{I} \). Standard assumptions necessary for effective interference suppression are that the desired symbol is uncorrelated with the interference and noise: \( \mathbb{E}[\mathbf{s}^\text{H} \mathbf{w}_I] = \mathbb{E}[\mathbf{s}^\text{H} \mathbf{w}_N] = 0 \). We also assume that the interference and noise are uncorrelated.

A linear correlator \( \mathbf{c} \) produces a decision statistic \( \mathbf{c}^\text{H} \mathbf{r} \) for the desired symbol, and its SINR is given by

\[
\text{SINR}(\mathbf{c}) = \frac{\mathbb{E}[|\mathbf{c}^\text{H} \mathbf{r}|^2]}{\mathbb{E}[|\mathbf{c}^\text{H} (\mathbf{w}_I + \mathbf{w}_N)|^2]} = \frac{\sigma_I^2 |\mathbf{c}^\text{H} \mathbf{h}|^2}{\mathbf{c}^\text{H} \mathbf{R}_I \mathbf{c} + \sigma_N^2 |\mathbf{c}|^2},
\]

where \( \mathbf{R}_I = \mathbb{E}[\mathbf{w}_I \mathbf{w}_I^\text{H}] \) is the interference covariance matrix, and \( \mathbf{R}_N = \mathbb{E}[\mathbf{w}_N \mathbf{w}_I^\text{H}] = \sigma_N^2 \mathbf{I} \) is the noise covariance matrix.

The LMMSE correlator minimizes \( \text{MSE} = \mathbb{E}[|\mathbf{c}^\text{H} \mathbf{r} - \mathbf{s}|^2] \) and maximizes \( \text{SINR} \) [35]. For the additive noise-plus-interference model (44), it is known to be proportional to a whitened matched filter (i.e., it suppresses interference by whitening it):

\[
\mathbf{c}_{\text{MMSE}} = \alpha(\mathbf{R}_I + \mathbf{R}_N)^{-1} \mathbf{h} = \alpha(\mathbf{R}_I + \sigma_N^2 \mathbf{I})^{-1} \mathbf{h},
\]

where \( \alpha \) is a scale factor that can be solved for easily (e.g., see [35]). Since SINR does not depend on scale factor, it is easy to show, plugging into (45), that

\[
\text{SINR} = \sigma_N^2 \mathbf{h}^\text{H} (\mathbf{R}_I + \mathbf{R}_N)^{-1} \mathbf{h} = \sigma_N^2 \mathbf{h}^\text{H} (\mathbf{R}_I + \sigma_N^2 \mathbf{I})^{-1} \mathbf{h}.
\]

Let us also for reference define the SNR:

\[
\text{SNR} = \sigma_N^2 \mathbf{h}^\text{H} (\mathbf{R}_N)^{-1} \mathbf{h} = \sigma_N^2 ||\mathbf{h}||^2 / \sigma_N^2.
\]

Remark 1. A positive definite matrix \( \mathbf{A}(\theta) \) increases with \( \theta \) if \( \mathbf{A}(\theta) - \mathbf{A}(\theta') \geq 0 \) for any \( \theta > \theta' \). That is, for any vector \( \mathbf{u} \), \( \mathbf{u}^\text{H} \mathbf{A}(\theta) \mathbf{u} \geq \mathbf{u}^\text{H} \mathbf{A}(\theta') \mathbf{u} \).

We can now infer the following properties relevant for our approach to performance analysis, stated as a lemma.

Lemma C.1. If the noise level \( \sigma_N^2 \) increases, with the signal and interference characteristics unchanged, then

(a) Absolute performance gets worse, with SINR and SNR both decreasing.

(b) The noise enhancement gets better: \( \frac{\text{SNR}}{\text{SNR}} \) decreases.

Proof: For (a), we note that the positive definite matrix \( \mathbf{R}_I + \sigma_N^2 \mathbf{I} \) increases with \( \sigma_N^2 \), hence its inverse decreases with \( \sigma_N^2 \). For (b), note that

\[
\frac{\text{SNR}}{\text{SNR}} = \frac{||\mathbf{h}||^2 / \sigma_N^2}{\mathbf{h}^\text{H} (\mathbf{R}_I + \sigma_N^2 \mathbf{I})^{-1} \mathbf{h}} = \frac{||\mathbf{h}||^2}{\mathbf{h}^\text{H} (\mathbf{R}_I + \sigma_N^2 \mathbf{I} + \mathbf{I})^{-1} \mathbf{h}}.
\]

The positive definite matrix \( \mathbf{R}_I + \sigma_N^2 \mathbf{I} + \mathbf{I} \) decreases with \( \sigma_N^2 \), hence its inverse increases with \( \sigma_N^2 \). Thus, the denominator on the right-hand side of equation (49) increases with \( \sigma_N^2 \), while the numerator is independent of it, proving the desired result.

ACKNOWLEDGMENTS

This work was supported in part by the Semiconductor Research Corporation (SRC) under the JUMP program (2018-JU-2778) and by DARPA (HR0011-18-3-0004). Use was made of the computational facilities administered by the Center for Scientific Computing at the CNSI and MRL (an NSF MRSEC; DMR-1720256) and purchased through NSF CNS-1725797.

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