Random Access Channel Coding in the Finite Blocklength Regime

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Abstract—The paper considers a random access communication scenario with an unknown number of transmitters. The collection of active transmitters remains fixed during each epoch, but neither the transmitters nor the receiver know which subset of transmitters that is nor even the number of active transmitters. All transmitters employ identical encoders. The channel is assumed to be defined for any number of possible transmitters and invariant to permutations on its inputs. The decoder is tasked with decoding from the channel output the number of active transmitters (\(k\)) and their messages but not with determining which transmitter sent which message; the decoding procedure occurs at a time \(n_k\) dependent on its estimate \(\hat{k}\) of the number of active transmitters, thereby achieving a rate that varies with the number of active transmitters. Orthogonalization strategies simplify coding by scheduling communications. The central result of this work demonstrates the achievability of performance that is first-order optimal for the MAC in operation during each coding epoch. Unlike prior multiple access schemes for a fixed number of transmitters requiring \(2^k - 1\) simultaneous threshold rules, the proposed scheme uses a single threshold rule and is optimal in terms of dispersion.

I. INTRODUCTION

Access points like WiFi hot spots and cellular base stations are, for wireless devices, the gateway to the network. Unfortunately, access points are also the network’s most critical bottleneck. As more kinds of devices become network-reliant, both the number of communicating devices and the diversity of their communication needs grow. Little is known about how to code under high variation in the number and variety of communicators.

Multiple-transmitter channels are well understood in information theory only when the number and identity of transmitters are fixed and known. Even in this known-transmitter regime, information-theoretic solutions are prohibitively complex to implement. As a result, orthogonalization methods, such as TDMA, FDMA, or orthogonal CDMA are used instead. Orthogonalization strategies simplify coding by scheduling the transmitters, but such methods can at best attain the single-transmitter capacity of the channel, which is significantly smaller than the multi-transmitter capacity. As a result, most random access protocols currently in use rely on collision avoidance, which again cannot surpass the single-transmitter capacity of the channel and may be significantly worse since the unknown transmitter set makes it difficult to schedule or coordinate among transmitters. Collision avoidance is achieved either through variations of the legacy (slotted) ALOHA or carrier sense multiple access (CSMA). ALOHA, which uses random transmission times and back-off schedules, achieves only \(\approx 37\%\) of the single-transmitter capacity of the channel [1]. In CSMA, each transmitter tries to avoid collisions by verifying the absence of other traffic before starting a transmission over the shared channel; when collisions do occur, for example because two transmitters begin transmission at the same time, they are handled by aborting the transmission, sending a jamming signal to be sure that all transmitters are aware of the collision, and then restarting the procedure at a random time, which again introduces inefficiencies. The state of the art in random access coding is “treating interference as noise,” which is part of newer CDMA-based standards. While this strategy can deal with random access better than ALOHA, it is still far inferior to the theoretical limits.

Even from a purely theoretical perspective, a satisfactory solution to random access remains to be found. The MAC model in which \(k\) out of \(K\) transmitters are active was studied by D’yachkov-Rykov [2] and Mathys [3] for zero-error coding on a noiseless adder MAC and Bassalygo and Pinsker [4] for an asynchronous model in which the information is considered erased if more than one transmitter is active at a time. See [5] for more. While zero-error code designs are mathematically elegant, they are also combinatorial in nature, and their complexity scales exponentially with the number of transmitters. Two-layer MAC decoders, with outer layer codes that work to remove channel noise and inner layer codes that work to resolve conflicts, are proposed in [6], [7]. Like the codes in [2], [3], the codes in [4], [6] are designed for a predetermined number of transmitters, \(k\); it is not clear how robust they are to randomness in the transmitters’ arrivals and departures. Minero et al. [8] studied a random access model in which the receiver knows the transmitter activity pattern, and the transmitters opportunistically send data at the highest possible rate. The receiver recovers only a portion of the messages sent, depending on the current level of activity in the channel.

This paper poses the question of whether it is possible, in a scenario where no one knows how many transmitters are active, for the receiver to almost always recover the messages sent by all active transmitters. Surprisingly, we find that not

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only is reliable decoding possible in this regime, but, for the class of permutation-invariant channels [5], it is possible to attain both the capacity and the dispersion of the MAC in operation; that is, we do as well in first- and second-order performance as if the transmitter activity were known everywhere a priori. Since the capacity region of a MAC varies with the number of transmitters, it is tempting to believe that the transmitters of a random access system must somehow vary their codebook size in order to match their transmission rate to the capacity region of the MAC in operation. Instead, we here allow the decoder to vary its decoding time depending on the observed channel output – thereby adjusting the rate at which each transmitter communicates by changing not the size but the blocklength of each transmitter’s codebook.

Codes that can accommodate variable decoding times are called \textit{rateless codes}. They were originally analyzed by Burnashev [9], who computed the error exponent of variable-length coding over a known point-to-point channel. Polanskiy et al. [10] provided a dispersion-style analysis of the same scenario. A practical implementation of rateless codes for an erasure channel with an unknown erasure probability appeared in [11]. An analysis of rateless coding over an unknown binary symmetric channel appeared in [12] and was extended to an arbitrary discrete memoryless channel in [13], [14] using a decoder that tracks Goppa’s empirical mutual information and decodes once that quantity passes a threshold. In [15], Jeffrey’s prior is used to weight unknown channels.

Unlike the codes described in [9]–[15], which allow truly arbitrary decoding times, in this paper we allow decoding only at a predetermined list of possible times $n_1, n_2, \ldots$. This strategy both eases practical implementation and reduces feedback. In particular, the schemes in [9]–[15] transmit a single-bit acknowledgment message from the decoder to the encoder(s) once the decoder is ready to decode. Because the decoding time is random, this so-called “single-bit” feedback forces the transmitter(s) to listen to the channel constantly, at every time step trying to discern whether or not a transmission was received, which requires full-duplex devices or doubles the effective blocklength and can be quite expensive. Thus while the receiver technically sends only “one bit” of feedback, the transmitters receive one bit of feedback (with the alphabet \{“transmit”,”no transmit”\}) in every time step, giving a feedback rate of 1 bit per channel use rather than a total of 1 bit. In our framework, acknowledgment bits are sent only at times $n_1, n_2, \ldots$, and thus the transmitters must tune in only occasionally.

In this paper, we view the random access channel as a collection of all possible MACs that might arise as a result of the transmitter activity pattern. Barring the intricacies of multiuser decoding, viewing an unknown channel as a collection of possible channels, without assigning an a priori probability to each, is known as the \textit{compound channel} model [16]. In the context of single-transmitter compound channels, it is known that if the decoding time is fixed, the transmission rate cannot exceed the capacity of the weakest channel from the collection [16], while the dispersion may be better (smaller) [17]. With feedback and allowing a variable decoding time, one can do much better [12]–[15].

Recently, Polanskiy [5] argued for removing the transmitter identification task from the physical layer encoding and decoding procedures. As he pointed out, such a scenario was previously discussed by Berger [18] in the context of conflict resolution. Polanskiy further suggested studying MACs whose conditional channel output distributions are insensitive to input permutations. For such channels, provided that all transmitters use the same codebook, the receiver can at best hope to recover the messages sent, but not the transmitter identity.

In this paper, we build a random access communication model from a family of such permutation-invariant MACs and employ identity-blind decoding at the receiver. Although not critical for the feasibility of our approach, these assumptions lead to a number of pleasing simplifications of both our scheme and its analysis. For example, the collection of MACs comprising our compound random access channel model can be parameterized by the number of active transmitters, rather than by the full transmitter activity pattern. If the maximum number of transmitters is finite, the analysis of identity-blind decoding differs little from traditional analyses that use independent realizations of a random codebook at each transmitter.

We provide a second-order analysis of the rate achieved by our multiuser scheme universally over all transmitter activity patterns, taking into account the possibility that the decoder may misdetect the current activity pattern and decode for a wrong channel. Leveraging our observation that for a symmetric MAC, the fair rate point is not a corner point of the capacity region, we are able to show that a single-threshold decoding rule attains the fair rate point. This differs significantly from traditional MAC analyses, in which $2^k - 1$ simultaneous threshold rules are used. In the context of a MAC with a known number of transmitters, second-order analyses of multiple-threshold decoding rules were obtained in [19]–[22] (finite alphabet MAC), and in [23] (Gaussian MAC). A non-asymptotic analysis of variable-length coding with single-bit feedback over a (known) Gaussian MAC was given in [24].

Other relevant recent works on MAC include the following. To account for massive numbers of transmitters, Chen and Guo [25], [26] introduced a notion of capacity for the multiple access scenario in which the maximal number of transmitters grows with blocklength and an unknown subset of transmitters is active at a given time. They show that time sharing, which achieves conventional MAC capacity, is inadequate to achieve capacity in that regime. On the effect of limited feedback on capacity of MAC, Sarwate and Gastpar showed in [27] that rate-0 feedback does not increase the no-feedback capacity of the discrete memoryless MAC whereas in compound MACs, it is possible to increase the capacity with a limited feedback by using a simple training phase to estimate the channel state.

In short, this paper develops a random access architecture with theoretical performance guarantees that can handle uncoordinated transmissions of a large and random number of
transmitters. Our system model and the proposed communication strategy are laid out in Section II. The main result is presented in Section III. The proofs are found in Section IV.

II. PROBLEM SETUP

For any positive integers \(i, j\) let \([i] = \{1, \ldots, i\}\) and \([i : j] = \{i, \ldots, j\}\), giving \([i : j] = \emptyset\) when \(i > j\). For any sequence \(x = (x_1, x_2, \ldots)\) and any ordered set \(C \subseteq \mathbb{N}\), vector \(x_C \triangleq (x_c: c \in C)\). For any vectors \(a, b\) with the same dimension, we write \(a \equiv b\) if and only if there exists a permutation \(\pi\) of \(b\) such that \(a = \pi(b)\).

A memoryless symmetric random access channel (henceforth called simply a RAC) is a memoryless channel with 1 receiver and an unknown number of transmitters. It is described by a family of stationary, memoryless MACs

\[
\left\{ \left( X^k, P_{Y_k|X_k} (y_k|x_k^{[k]}), Y_k \right) \right\}_{k=1}^K
\]

(1)

each indexed by a number of transmitters, \(k\); the maximal number of transmitters is \(K\) for some \(K \leq \infty\). The \(k\)-transmitter MAC has input alphabet \(X^k\), output alphabet \(Y_k\), and conditional distribution \(P_{Y_k|X_k}\). When \(k\) transmitters are active, the RAC output is \(Y = Y_k\). To capture the property that the impact of a channel input on the channel output is independent of the transmitter from which it comes, each channel in (1) is assumed to be permutation-invariant; that is,

\[
P_{Y_k|X_k} (y_k|x_k^{[k]}) = P_{Y_{\pi(k)}|X_k} (y_k|x_{\pi([k])})
\]

\forall \text{ permutation } \pi \text{ on } [k].

(2)

Since, for any \(s < k\), MAC-s is physically identical to MAC-k operated with \(s\) active and \(k - s\) silent transmitters, we use \(0 \in X\) to represent transmitter silence and require reducibility:

\[
P_{Y_{[s]}|X_{[s]}} (y|x_{[s]}) = P_{Y_k|X_k} (y|x_k^{[k]})^s (0^{k-s}) \quad \forall s < k
\]

(3)

for all \(x_{[s]} \in X_{[s]}\) and \(y \in Y_s\). An immediate consequence of this notion of reducibility is that \(Y_s \subseteq Y_k\) for any \(s < k\).

We here propose a new RAC communication strategy. In the proposed strategy, communication occurs in epochs, with each epoch beginning in the time step following the previous epoch’s end. Each epoch ends with a single acknowledgment bit (ACK), which the receiver broadcasts to all transmitters as described below. At the beginning of each epoch, each transmitter independently decides whether to be active or silent in that epoch; the decision is binding for the length of the epoch, meaning that a transmitter must either actively transmit for all time steps in the epoch or remain silent for the same period. Thus while the total number of transmitters is potentially unlimited, the number of active transmitters, \(k\), stays constant during the entire transmission period between two ACKs.

Each transmitter uses the epoch to describe a message \(W\) from the alphabet \([M]\); when the active transmitters are \([k]\), the messages \(W_{[k]} \in [M]^k\) are independent and uniformly distributed. The receiver considers decoding at each time \(n_1, n_2, \ldots\), choosing to decode at time \(n_t\) only if it believes at that time that the number of active transmitters is \(t\). The transmitters are informed of the decoder’s decision about when to stop transmitting by a single-bit acknowledgment (ACK) \(Z_i\) broadcasted at each time \(n_i\) with \(i \in [t]\); here \(Z_i = 0\) for all \(i < t\) and \(Z_i = 1\), with “1” signaling the end of one epoch and the beginning of the next.

It is important to stress that in this domain, each transmitter knows nothing about the set of active transmitters \(A \subset \mathbb{N}\) beyond its own membership and what it learns from the receiver’s feedback, and the receiver knows nothing about \(A\) beyond what it learns from the channel output \(Y\). (We call this agnostic random access.) In addition, since designing a different encoder for each transmitter is expensive from the perspective of both code design and code operation, as in [5], we assume that every transmitter employs the same encoder. (We call this symmetrical encoding.) Under these assumptions, what the transmitters and receiver can learn about \(A\) is quite limited. In particular, the reducibility, permutation invariance, and symmetrical encoding properties together imply that the decoder can at best hope to distinguish which messages were transmitted rather than by whom they were sent. In practice, transmitter identity could be included in the header of each log \(M\)-bit message or at some other layer of the stack; transmitter identity is not, however, handled by the RAC code. Instead, since the channel output statistics depend on the dimension of the channel input but not the identity of the active transmitters, the receiver’s task is to decode the messages transmitted but not the identities of their senders. We therefore assume without loss of generality that \(|A| = k\) implies \(A = [k]\), and thus the family of \(k\)-transmitter MACs in (2) indeed fully describes the behavior of a RAC.

The single-bit feedback strategy described above allows us to use rateless codes to deal with the agnostic nature of random access. Specifically, prior to the transmission, the decoder fixes the blocklengths \(n_1 < n_2 < \ldots\), where \(n_t\) is the decoding blocklength when the decoder decides that the number of active transmitters \(k\) is equal to \(t\). As we show in Section IV below, with an appropriately designed decoding rule, with high probability, correct decoding is performed at time \(n_k\). Naturally, the greater the number of active transmitters the longer it takes to decode. The following definition formalizes such rateless codes for agnostic random access.

**Definition 1.** An \((M, \{(n_k, \epsilon_k)\})_{k=1}^K\) code for a RAC is the (rateless) encoding function

\[
f: [M] \rightarrow X^{n_k}
\]

(4)

and a collection of decoding functions:

\[
g_k: Y^{n_k} \rightarrow [M] \cup \epsilon, \quad k = 1, 2, \ldots, K
\]

(5)

such that if \(k\) transmitters are active, then, with probability at least \(1 - \epsilon_k\), the \(k\) messages are correctly decoded at time \(n_k\).

\[1\]The maximum number of transmitters \(K = +\infty\) is permitted, in which case \(n_{k+1}\) in (4) is replaced by \(\infty\).
That is,2
\[ \frac{1}{M^k} \sum_{\mathcal{L} \in [M]^k} \mathbb{P}[g_k(Y^n_k) \neq \mathcal{L}|W_k = \mathcal{L}] \leq \epsilon_k, \quad (6) \]
where \( W_k \) are the transmitters’ messages, independent and equiprobable on \([M], X^n_a = f(W_a)_{a=1}^k, s = 1, \ldots, k, \) and
\[ P_{Y^k|X^n_a} = P_{Y^n_k|X^n_s}. \]

Under symmetrical encoding, each transmitter uses the same encoder, \( f, \) to form a codeword of length \( n_K \) (which might be \( +\infty \)), which is fed into the channel symbol-by-symbol. According to Definition 1, if \( k \) transmitters are active then the decoder recovers the sent messages correctly after observing the first \( n_k \) channel outputs, with probability at least \( 1 - \epsilon_k \). The decoder \( g_k \) does not attempt to recover transmitter identity; successful decoding means that the list of messages it outputs coincides with the list of messages sent.

The following definitions are useful for the discussion that follows. When \( k \) transmitters are active, marginal distribution \( P_{Y_k} \) is determined by the input distribution \( P_{X^k} \). The information density and conditional information density are then defined as
\[ i_k(x_A; y_k) \triangleq \log \frac{P_{Y_k|x_A}(y_k|x_A)}{P_{Y_k}(y_k)} \]
\[ i_k(x_A; y_k|x_B) \triangleq \log \frac{P_{Y_k|x_A,x_B}(y_k|x_A,x_B)}{P_{Y_k|x_B}(y_k|x_B)} \]
for any \( A, B \subseteq \{1, \ldots, k\} \); here \( i_k(x_A; y_k|x_B) \triangleq i_k(x_A; y_k) \) when \( B = \emptyset \) and \( i_k(x_A; y_k|x_B) \triangleq 0 \) when \( y_k \notin Y_k \) or \( x_A = \emptyset \). The corresponding mutual informations are
\[ I_k(x_A; Y_k) \triangleq \mathbb{E}[i_k(x_A; Y_k)], \quad (9) \]
\[ I_k(x_A; Y_k|x_B) \triangleq \mathbb{E}[i_k(x_A; Y_k|x_B)]. \quad (10) \]
Throughout, we also denote for brevity
\[ I_k \triangleq I_k(X_k[Y_k]), \quad (11) \]
\[ V_k \triangleq \text{Var}[i_k(X_k[Y_k])]. \quad (12) \]

To ensure the existence of codes satisfying the error constraints in Definition 1, we assume that there exists a \( P_X \) such that when \( X_1, X_2, \ldots, X_K \) are distributed i.i.d. \( P_X \), then the conditions in (13)–(17) below are satisfied.

The friendliness assumption states that for all \( k \leq K, \)
\[ I_k(X_s[Y_k]; X_{s+1:k}) = 0^{k-s} \geq I_k(X_s[Y_k]; X_{s+1:k}). \quad (13) \]
Friendliness implies that a transmitter that remains silent is at least as good from the perspective of the decoder as a transmitter that reveals its transmission to the decoder. Naturally, (13) can always be satisfied with an appropriate designation of the “silence” symbol.

Next, the interference assumption states that \( X_s \) and \( X_{s+1:k} \) are conditionally dependent given \( Y_k \) for any \( 1 \leq s < t \leq k, \) i.e.
\[ P_{X_t|Y_k} \neq P_{X_t|Y_k} \cdot P_{X_{t+1:k}|Y_k} \quad \forall 1 \leq s < t \leq k, \forall k. \quad (14) \]
2Recall that \( = / \neq \) denote equality/inequality up to a permutation.

This interference assumption (14) eliminates a trivial RAC in which there is no interference between different transmitters.

Finally, the following moment assumptions enable the second-order analysis presented in Theorem 1 below:
\[ \text{Var}[i_k(X_k[Y_k])], \quad (15) \]
\[ \mathbb{E}[|i_k(X_k[Y_k]) - I_k(X_k[Y_k])|^2] < \infty, \quad (16) \]
\[ \text{Var}[i_t(X_s[Y_k]), \quad \forall s \leq t \leq k. \quad (17) \]

All discrete memoryless channels (DMCs) satisfy (16)–(17) [28, Lemma 46] as do Gaussian noise channels. Further, common channel models from the literature typically satisfy (15) as well.

For example, channels meeting (2), (3), (13), (14), (15)–(17) include the AWGN RAC,
\[ Y_k = k \sum_{i=1}^{k} X_i + Z, \quad (18) \]
where \( X \in \mathbb{R} \) operates under a power constraint \( P \) and \( Z \sim \mathcal{N}(0, \sigma^2) \) for some \( \sigma^2 > 0, \) and the adder-erasure RAC,
\[ Y_k = \begin{cases} \sum_{i=1}^{k} X_i, & \text{w.p. } 1 - \delta \\ e, & \text{w.p. } \delta, \end{cases} \quad (19) \]
where \( X_i \in \{0, 1\}, \) \( Y_k \in \{0, \ldots, k\} \cup e. \)

We conclude this section with a series of lemmata that describe the natural orderings possessed by the collection of channels in (1). These properties are key to the feasibility of our achievability scheme, presented in the next section. They are a consequence of our assumptions in (2), (3) (13), and (14). Proofs are relegated to the Appendix.

The first lemma describes a natural property of the collection of channels in (1): the quality of the channel for each transmitter deteriorates as more transmitters are added (even though the sum capacity increases).

**Lemma 1.** Let \( X_1, X_2, \ldots, X_k \) be i.i.d. Under permutation invariance (2), reducibility (3), friendliness (13), and interference (14),
\[ I_k < I_s \quad \text{for } s < k. \quad (20) \]
Furthermore, the following inequalities hold.

**Lemma 2.** Let \( X_1, X_2, \ldots, X_k \) be i.i.d. Under permutation invariance (2), reducibility (3) and interference (14), it holds for all \( s < k, \)
\[ \frac{1}{k} I_k(X_k[Y_k]) < \frac{1}{s} I_k(X_s[Y_k]; X_{s+1:k}). \quad (21) \]

**Lemma 3.** Let \( X_1, X_2, \ldots, X_k \) be i.i.d. If a RAC is permutation-invariant (2), reducible (3), friendly (13), and exhibits interference (14), then for any \( s \leq t < k, \) we have
\[ \mathbb{E}[i_t(X_s[Y_k])] \leq I_k(X_s[Y_k]) < I_s(X_s[Y_t]). \quad (22) \]
Theorem 1 bounds the performance of a finite blocklength RAC code. For any number $k$ of active transmitters, the code achieves a rate vector $R_{[k]} = (R_1, \ldots, R_k)$, with sum rate $kR$ converging as $O\left(\frac{1}{\sqrt{n_k}}\right)$ to $I_k(X_{[k]}; Y_k)$ for some input distribution $P_{X_{[k]}|Y_k}(x_{[k]}|y_k) = \prod_{i=1}^{k} P_x(x_i)$ with $P_x(x)$ independent of $k$. Thus for any family of MACs for which a single $P_x$ maximizes $I_k(X_{[k]}; Y_k)$ for all $k$, the proposed sequence of codes converges to the symmetrical rate point on the capacity region of the MAC with the same number of transmitters.\(^3\)

**Theorem 1. (Achievability) For any RAC**

\[
\left\{\left(X^k, P_{Y_k|X_{[k]}}(y_k|x_{[k]}), Y_k\right)\right\}_{k=1}^{K}
\]

satisfying (2), (3), any $K < \infty$ and any fixed $P_X$ satisfying (13)–(17), there exists an $(M, \{(n_k, \epsilon_k)\}_{k=1}^{K})$ code provided

\[
\log M \leq \frac{1}{k} \left\{ n_k I_k - \sqrt{n_k} k Q^{-1}(\epsilon_k) - \frac{1}{2} \log n_k + O(1) \right\},
\]

for all $k \leq K$, where $Q(x) = \frac{1}{\sqrt{2\pi}} \int_{x}^{\infty} \exp \left\{ -\frac{u^2}{2} \right\} du$ is the Gaussian complementary cumulative distribution function.

To shed light on the statement of Theorem 1, suppose that the channel is such that the same distribution $P_X$ satisfying (13)–(17) achieves the maximum of $I_k$, for all $k$. For example, for the add-erasure RAC in (19), Bernoulli-1/2 $P_X$ attains $\max I_k, V_k$. Thanks to Lemma 1, for $M$ large enough and any $\epsilon_1, \epsilon_2, \ldots, \epsilon_K$, one can pick $n_1 < n_2 < \ldots < n_K$ so that the right side of (23) is equal to $\log M$, for all $k$. Therefore, somewhat counter-intuitively, Theorem 1 certifies that using rateless codes with acknowledgments, it is in some cases possible to transmit over the RAC using a transmission scheme that is completely agnostic to the transmitter activity pattern and to perform as well (in terms of both first and second order terms in (23)) as the optimal transmission scheme designed with complete knowledge of transmitter activity.

Theorem 1 follows by an application of Theorem 2, which bounds the error probability of the finite-blocklength RAC code defined in Section IV. When $k$ transmitters are active, the error probability is $\epsilon_k$, which captures both errors in the estimate $t$ of $k$ and errors in the reproduction $W_t$ of $W_k$ when $t = k$.

**Theorem 2. For any RAC**

\[
\left\{\left(X^k, P_{Y_k|X_{[k]}}(y_k|x_{[k]}), Y_k\right)\right\}_{k=1}^{K}
\]

satisfying (2), (3), any $K < \infty$ and a fixed input distribution $P_X$, there exists an $(M, \{(n_k, \epsilon_k)\}_{k=1}^{K})$ code with

\[
\epsilon_k \leq \mathbb{P}[t_k(X_{[k]}^{n_k}; Y_{k}^{n_k}) \leq \log \gamma_k]
\]

\[
+ \sum_{i=1}^{k-1} \left( k \right) \mathbb{P}[t_k(X_{[k]}^{n_k}; Y_{k}^{n_k}) > \log \gamma_i]
\]

\[
+ \sum_{i=1}^{k} \frac{k}{t} \left( \mathbb{P}[t_k(X_{[k]}^{n_k}; Y_{k}^{n_k}) > \log \gamma_i] - \frac{M}{k} \right)
\]

\[
\text{for all } k, \text{ where } \lambda_{s,t}^k, \text{ and } \gamma_i \text{ values are arbitrary constants and, for any } n, \ (X_{[k]}^{n_k}, X_{[k]}^{n_k}; Y_{[k]}^n) \text{ represents a random sequence drawn i.i.d. according to } P_{X_{[k]}|Y_{[k]}}(x_{[k]}|y_k, x_{[k]}; y_k).
\]

In the operational regime of interest, the dominating term is (24), which is the probability that the true codeword set produces a low information density. The remaining terms are all negligible, as seen in the refined asymptotic analysis of the bound in Theorem 2 (see Section IV-B, below). The remaining terms bound the probability that two or more transmitters pick the same codeword (25), the probability that the decoder estimates the number of active transmitters as $t$ for some $t < k$ and decodes those $t$ messages correctly (26), and the probability that the decoder estimates the number of active transmitters as $t$ for some $t < k$ and decodes the messages from $s$ of those $t$ transmitters incorrectly and the messages from the remaining $t - s$ of those transmitters correctly (27)–(28).

**Encoder Design:** As described in Section II, an $(M, \{(n_k, \epsilon_k)\}_{k=1}^{K})$ RAC code employs the same encoder $f(x)$ at every transmitter. For any $u_{[k]} \in [M]^{k}$, we use...
For each $\gamma_g$ for some fixed $P_X$ for some fixed $X$, decoder $g_k$ employs a single threshold rule

$$g_k(y_{nk}) = \begin{cases} \bar{w}[k] & \text{if } t_k(f(w_{nk} | m_k); y_{nk}) > \log \gamma_k \\
 \text{e otherwise} \end{cases}$$

for some constant $\gamma_k$, chosen before the transmission starts. By permutation-invariance (2) and symmetrical encoding, all permutations of the message vector $w_{nk}$ give the same mutual information density. We use the ordered permutation specified in (37) as a representative of the equivalence class with respect to the binary relation $\equiv$. The choice of a representative is immaterial since decoding is identity-blind.

When there is more than one ordered $w_{nk}$ that satisfies the threshold condition, decoder $g_k$ chooses among them uniformly at random. All such events are included in the error probability bound below.

The proof of Theorem 2, below, bounds the error probability of the proposed code.

### A. Proof of Theorem 2

In the discussion that follows, we bound the error probability of the code $(f, \{g_k\}_{k=1}^K)$ defined above. The core of the analysis relies on the independence of codewords $f(w_i)$ and $f(W_j)$ from distinct transmitters $i$ and $j$. By the given code design, this assumption is valid provided that $W_i \neq W_j$; we therefore count events of the form $W_i = W_j$ among our error events. Let $P_{\text{rep}}$ denote the probability of such a repetition; then

$$P_{\text{rep}} = 1 - \prod_{i=0}^{k-1} (M - i) / M^k. \quad (38)$$

The discussion that follows uses $w_{nk}^* = (1, \ldots, k)$ as an example instance of a message vector $w_{nk}$ in which $w_i \neq w_j$ for all $i, j$ and, $Y_1[n]$ as the set of all message vectors $\bar{w}[s]$ for which $\bar{w}_i \neq \bar{w}_j$ for all $i \in [s], j \neq \bar{w}_j$ for all $i < j \in [s]$, giving $Y_1[n] = \{\bar{w}[s] \in [M]^n : \bar{w}_i > k, \bar{w}_j < \bar{w}_j \forall i < j\}$. Note that we need to include only ordered vectors in $Y_1[n]$ in view of our identity-blind decoding rule in (37). The resulting error bound proceeds as (39)–(44), displayed at the top of the next page, where $X[k]$ is the vector of transmitted codewords and $\tilde{X}[s](\bar{w}[s])$ represents the codeword for $\bar{w}[s]$,

which was not transmitted. Line (40) separates the case where distinct transmitters send the same message from the case where there is no repetition. Lines (41)–(42) enumerate the error events in the no-repetition case; these include all cases where the transmitted codeword fails to meet the threshold (41), all cases where a prefix of the transmitted codeword meets the threshold for some $t < k$ (41), and all case where a codeword that is wrong in $s$ dimensions and right in $t - s$ dimensions meets the threshold for $t \leq k$ (42). We apply the union bound and the symmetry of the code design to represent the probability of each case by the probability of an example instance times the number of instances. Equations (43)–(44) replace decoders by the threshold rules in their definitions. The delay in applying the union bound in the final line is deliberate. Applying the following observation before applying the union bound yields a tighter bound.

$$\begin{align*}
P & \left[ \bigcup_{\bar{w}[s] \in Y_1[n]} \{t(X_{nk}[\bar{w}[s]], X_{nk}[s+1:t]; Y_k) > \log \gamma_t\} \right] \\
& = P \left[ \left( \bigcup_{\bar{w}[s] \in Y_1[n]} \{t(X_{nk}[\bar{w}[s]], X_{nk}[s+1:t]; Y_k) > \log \gamma_t\} \right) \\
& \cap \{t(X_{nk}[s+1:t]; Y_k) > n_k \mathbb{E}[t(X_{nk}[s+1:t]; Y_k)] + \lambda_k\} \right] \\
& < P \left[ \left( \bigcup_{\bar{w}[s] \in Y_1[n]} \{t(X_{nk}[\bar{w}[s]], X_{nk}[s+1:t]; Y_k) > \log \gamma_t\} \right) \\
& \cap \{t(X_{nk}[s+1:t]; Y_k) > n_k \mathbb{E}[t(X_{nk}[s+1:t]; Y_k)] + \lambda_k\} \right] \\
& \leq P \left[ \{t(X_{nk}[s+1:t]; Y_k) > n_k \mathbb{E}[t(X_{nk}[s+1:t]; Y_k)] + \lambda_k\} \right] \\
& + P \left[ \bigcup_{\bar{w}[s] \in Y_1[n]} \{t(X_{nk}[\bar{w}[s]]; Y_k) > \log \gamma_t - n_k \mathbb{E}[t(X_{nk}[s+1:t]; Y_k)] - \lambda_k\} \right] \Bigg]. \quad (46)$$

Therefore

$$\epsilon_k \leq P_{\text{rep}} + P \left[ t_k(X_{nk}[k]; Y_k) \leq \log \gamma_k \right]$$

$$+ \sum_{t=1}^{k-1} \binom{k}{t} P \left[ t_t(X_{nk}[s]; Y_k) > \log \gamma_t \right] \quad (47)$$

$$+ \sum_{i=1}^{k} \sum_{s=1}^{t-s} \binom{k}{s} P \left[ t_t(X_{nk}[s+1:t]; Y_k) > n_t \mathbb{E}[t_t(X_{nk}[s+1:t]; Y_k)] + \lambda_k\right]$$

$$+ \sum_{i=1}^{k} \sum_{s=1}^{t-s} \binom{k}{s} P \left[ t_t(X_{nk}[s+1:t]; Y_k) > \log \gamma_t - n_t \mathbb{E}[t_t(X_{nk}[s+1:t]; Y_k)] - \lambda_k\right], \quad (49)$$

which gives the desired result.
\[ e_k = \frac{1}{Mk} \sum_{w_{[k]} \in [M]^k} \mathbb{P}\left( \cup_{t=1}^{k-1} g_t(Y_k^{\ast t}) \neq e \cup g_k(Y_k^{\ast t}) \neq w_{[k]} | W_k = w_{[k]} \right) \]

\[ \leq \mathbb{P}_{\text{rep}} + \left( 1 - \mathbb{P}_{\text{rep}} \right) \mathbb{P}\left( \cup_{t=1}^{k-1} g_t(Y_k^{\ast t}) \neq e \cup g_k(Y_k^{\ast t}) \neq w_{[k]} | W_k = w_{[k]} \right) \]

\[ \leq \mathbb{P}_{\text{rep}} + \mathbb{P}[g_k(Y_k^{\ast n_k}) = e | W_k = w_{[k]}] + \sum_{t=1}^{k-1} \binom{k}{t} \mathbb{P}[g_t(Y_k^{\ast t}) \neq w_{[t]} | W_k = w_{[k]}] \]

\[ + \sum_{t=1}^{k} \sum_{s=1}^{t} \binom{k}{t-s} \mathbb{P}[\cup_{\tilde{w}_{[s]} \in \tilde{W}_{[s]}} (g_t(Y_k^{\ast s}) \neq (\tilde{w}_{[s]}, w_{[s+t-1]}) | W_k = w_{[k]}] \]

\[ \leq \mathbb{P}_{\text{rep}} + \mathbb{P}[t_k(X_k^{\ast n_k}; Y_k^{\ast n_k}) \leq \log \gamma_k] + \sum_{t=1}^{k-1} \binom{k}{t} \mathbb{P}[t_k(X_k^{\ast t}; Y_k^{\ast t}) > \log \gamma_t] \]

\[ + \sum_{t=1}^{k} \sum_{s=1}^{t} \binom{k}{t-s} \mathbb{P}[\cup_{\tilde{w}_{[s]} \in \tilde{W}_{[s]}} \{t_k(\tilde{X}_k^{\ast s}(\tilde{w}_{[s]}), X_k^{\ast s+t-1}; Y_k) > \log \gamma_t\}] \]

\[ \mathbb{P}[t_k(X_k^{\ast n_k}; Y_k^{\ast n_k}) > \log \gamma_t] \leq \frac{\text{Var}[t_k(X_k^{\ast t}; Y_k)]}{\text{Var}[t_k(X_k^{\ast t}; Y_k)] - \tau t \sqrt{\frac{M}{M}}}. \]

\[ \mathbb{P}[t_k(X_k^{\ast n_k}; Y_k^{\ast n_k}) > \log \gamma_t] \leq \frac{\text{Var}[t_k(X_k^{\ast t}; Y_k)]}{n_t \left( \frac{1}{2} \left( I_k(X_k^{\ast t}; Y_k) - t \sqrt{\frac{M}{M}} \right) \right)^2}. \]

\[ \mathbb{P}[t_k(X_k^{\ast n_k}; Y_k^{\ast n_k}) > \log \gamma_t] \leq \frac{\text{Var}[t_k(X_k^{\ast t}; Y_k)]}{n_t \left( \frac{1}{2} \left( I_k(X_k^{\ast t}; Y_k) - t \sqrt{\frac{M}{M}} \right) \right)^2}. \]

\[ \mathbb{P}[t_k(X_k^{\ast n_k}; Y_k^{\ast n_k}) > \log \gamma_t] \leq \frac{\text{Var}[t_k(X_k^{\ast t}; Y_k)]}{n_t \left( \frac{1}{2} \left( I_k(X_k^{\ast t}; Y_k) - t \sqrt{\frac{M}{M}} \right) \right)^2}. \]

\[ \mathbb{P}[t_k(X_k^{\ast n_k}; Y_k^{\ast n_k}) > \log \gamma_t] \leq \frac{\text{Var}[t_k(X_k^{\ast t}; Y_k)]}{n_t \left( \frac{1}{2} \left( I_k(X_k^{\ast t}; Y_k) - t \sqrt{\frac{M}{M}} \right) \right)^2}. \]
\[ \log \gamma_t - n_t I_t(X_{[s+1:t]};Y_t) - \lambda_{s,t}^k \geq \mathbb{E} \left[ \exp \left\{ t \sum_{i=s+1}^t \left( X_{[i]}^{n_i} - Y_{[i]}^{n_i} | X_{[s+1:t]}^{n_s} \right) \right\} \right] \cdot \exp \left\{ - \log \gamma_t - n_t I_t(X_{[s+1:t]};Y_t) - \lambda_{s,t}^k \right\} \]
\[ = \exp \left\{ - \log \gamma_t - n_t I_t(X_{[s+1:t]};Y_t) - \lambda_{s,t}^k \right\}. \]

Plugging our parameter choices (50), (51), (52) into (61) and using
\[ \left( \frac{u}{k} \right) \leq \left( \frac{en}{k} \right)^k, \]
we get,
\[ \left( M - k \right) \frac{P}{s} \left[ \sum_{i=s+1}^t \left( X_{[i]}^{n_i} - Y_{[i]}^{n_i} | X_{[s+1:t]}^{n_s} \right) \right] > \log \gamma_t - n_t I_t(X_{[s+1:t]};Y_t) - \lambda_{s,t}^k \]
\[ \leq \mathbb{E} \left[ \left. \exp \left\{ - n_t I_t(X_{[s+1:t]};Y_t) - s \cdot t / \gamma_t \right\} \right. \right] \cdot \left( 1 - s \right) \gamma_t \sqrt{n_t V_t} - \frac{s}{2} \log n_t + s \log \left( \frac{t}{s} \right). \]

Lemma 2 ensures that the exponent in (64) is negative for a large enough \( n_t \).
For \( s = t < k \), substituting the parameter choices (50), (51), (52) into (61) and using (62), we get,
\[ \left( M - k \right) \frac{P}{t} \left[ \sum_{i=t+1}^t \left( X_{[i]}^{n_i} - Y_{[i]}^{n_i} | X_{[s+1:t]}^{n_s} \right) \right] > \log \gamma_t \leq \frac{M-k}{\gamma_t} \leq \frac{1}{\sqrt{n_t}}. \]

Now suppose that \( s = t = k \). Following the change of measure technique (e.g. [30]), one can rewrite an expectation with respect to measure \( Q \) as an expectation with respect to measure \( P \) as follows:
\[ Q[Z \in A] = \mathbb{E}_P \left[ \left( \frac{dP}{dQ} (Z) \right)^{-1} \right]_{\{Z \in A\}}. \]

Switching to the measure \( P_{k \mid X[k]} P_{X(k)} \) in this way, by (62) and the parameter choice (52) we write
\[ \left( M - k \right) \frac{P}{k} \left[ \sum_{i=k+1}^t \left( X_{[i]}^{n_i} - Y_{[i]}^{n_i} | X_{[s+1:t]}^{n_s} \right) \right] \geq \log \gamma_k \]
\[ \leq \left( \frac{e}{k} \right)^k \left( M - k \right) \frac{P}{k} \left[ \sum_{i=k+1}^t \left( X_{[i]}^{n_i} - Y_{[i]}^{n_i} | X_{[s+1:t]}^{n_s} \right) \right] \cdot \mathbb{E} \left[ \exp \left\{ - t_k(X_{[k]}^{n_k} - Y_{[k]}^{n_k}) \right\} \mathbf{1}\{t_k(X_{[k]}^{n_k} - Y_{[k]}^{n_k}) > \log \gamma_k \} \right] \leq D_k / n_k, \]
where
\[ D_k \triangleq 2 \left( \frac{\log 2}{\sqrt{2 \pi \sqrt{k}}} + 2B_k \right). \]

To justify (68), notice that \( t_k(X_{[k]}^{n_k} - Y_{[k]}^{n_k}) \) is a sum of i.i.d. random variables; Polanskyi et al. in [28, Lemma 47] gave a sharp bound on \( \mathbb{E} \exp (- \sum_i Z_i) \mathbf{1}\{\sum_i Z_i > \gamma\} \), where the \( Z_i \)'s are independent. A direct application of that bound yields (68). Note that \( D_k \) is finite by the moment assumptions (15) and (16). Combining the bounds for the three cases in (64), (65) and (68), we conclude that (28) contributes \( O \left( \frac{1}{\sqrt{n_k}} \right) \) to the total error.

Finally, we set the constant \( C_k \) to ensure
\[ (25) + (26) + (27) + (28) \leq \frac{C_k}{\sqrt{n_k}}. \]

The existence of such a constant is ensured by our analysis above demonstrating that the terms (25)–(28) do not contribute more than \( O \left( \frac{1}{\sqrt{n_k}} \right) \) to the total.\(^5\)

Due to (54) and (70), the total probability of making an error at time \( n_k \) is bounded by \( \epsilon_k \), and in view of (53) the proof of Theorem 1 is complete.

\[ \square \]

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\(^5\)Our bounds on (25)–(28) technically depend on \( \gamma_k \) and therefore on \( C_k \); however it’s easy to see that their dependence on \( C_k \) is weak and for large enough \( n_k \) can be eliminated entirely; thus the choice of \( C_k \) satisfying (70) is possible.
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APPENDIX

We first state and prove Lemma 4, which we then use to prove Lemmas 2, 1 and 3 (in that order).

**Lemma 4** (Conditioning increases mutual information). Let $X_1, X_2, \ldots, X_k$ be independent, and let the interference assumption (14) hold. Then, for all $s < t \leq k$,

$$I_k(X_s; Y_k) < I_k(X_s; Y_k|X_{s+1:t}).$$  \hfill (71)

**Proof of Lemma 4.** By the chain rule for mutual information, when $U$ and $V$ are independent,

$$I(U; Y) \leq I(U; Y|V),$$  \hfill (72)

and (71) (with $\leq$ instead of $<$) follows by substituting $Y \leftarrow Y_k, U \leftarrow X_s, V \leftarrow X_{s+1:t}$ in (72). Equality in (72) is attained if and only if $U$ and $V$ are conditionally independent given $Y$. The interference assumption (14) eliminates the possibility of equality in (71). Q.E.D.

**Proof of Lemma 2.** Let $\ell = \lfloor k/s \rfloor$. By the chain rule of mutual information,

$$\frac{1}{k} I_k(X_k; Y_k) = \frac{1}{k} \sum_{i=1}^{\ell} I_k(X_{(i-1)s+1:is}; Y_k|X_{(i-1)s}) + \frac{1}{k} I_k(X_{[is+1:k]}; Y_k|X_{[is]}).$$  \hfill (73)

The first term in (73) is bounded using Lemma 4 as,

$$\frac{1}{k} \sum_{i=1}^{\ell} I_k(X_{[(i-1)s+1:is]}; Y_k|X_{(i-1)s}) \geq \frac{1}{k} I_k(X_s; Y_k)$$  \hfill (74)

where the inequality is strict for $\ell > 1$. To bound the second term in (73), we successively apply the chain rule of mutual information, Lemma 4 with permutation invariance (2), and chain rule again:

$$\frac{1}{k} I_k(X_{[is+1:k]}; Y_k|X_{[is]})$$

$$= \frac{1}{k} \sum_{i=1}^{\ell - is} I_k(X_{[is+i:is+1-k]}; Y_k|X_{[is+i-1]})$$

$$\geq \frac{k - is}{k} \sum_{i=1}^{s} I_k(X_s; Y_k|X_{[is-1]})$$

$$= \left(\frac{1}{s} - \frac{\ell}{k}\right) I_k(X_s; Y_k).$$  \hfill (77)

Combining (74) and (77), we obtain,

$$\frac{1}{k} I_k(X_k; Y_k) > \frac{1}{s} I_k(X_s; Y_k).$$  \hfill (78)

The inequality (78) is strict since for $l > 1$, (74) is strict and for $l = 1$, we have $k > ls$ and (77) is strict when $k > ls$ due to Lemma 4.

Substituting $s \leftarrow k - s$ in (78),

$$\frac{1}{k} I_k(X_k; Y_k) > \frac{1}{k-s} I_k(X_{k-s}; Y_k).$$  \hfill (79)

Rearranging, we get

$$k I_k(X_k; Y_k) > s I_k(X_s; Y_k) + k I_k(X_{k-s}; Y_k),$$  \hfill (80)

and (21) follows by subtracting $k I_k(X_{k-s}; Y_k)$ from both sides, using the chain rule, changing the labels by permutation invariance (2), and rearranging the terms again. \hfill Q.E.D.

**Proof of Lemma 1.**

$$I_s = I_k(X_s; Y_k|X_{[s+1:k]} = 0^{k-s})$$  \hfill (81)

$$\geq I_k(X_s; Y_k|X_{[s+1:k]})$$  \hfill (82)

$$> \frac{s}{k} I_k$$  \hfill (83)

where (81) is by reducibility (3), (82) is by friendliness (13), and (83) follows by Lemma 2. Q.E.D.

**Proof of Lemma 3.** We denote Kullback-Leibler divergence by $D(\cdot || \cdot)$, and we write

$$E[\ell_k(X_s; Y_k)] = -D(P_{Y_k|x_s}||P_{Y_k|x_s}) + D(P_{Y_k|x_s}||P_{Y_k}) + D(P_{Y_k}||P_{Y_k}).$$

$$= -D(P_{Y_k|x_s}||P_{Y_k|x_s}) + I_k(X_s; Y_k) + D(P_{Y_k}||P_{Y_k})$$

$$\leq I_k(X_s; Y_k)$$

$$< I_k(X_s; Y_k|X_{[s+1:s+k-\ell-1]})$$

$$= I_k(X_s; Y_k|X_{[s+1:k+\ell-1]})$$

(87)  \hfill (88)
\[ I_k(X_{[s]}; Y_k|X_{[t+1:k]} = 0^{k-t}) \geq I_k(X_{[s]}; Y_k) \]

where (86) follows from data processing inequality of relative entropy, (87) follows from Lemma 4 by substituting \( s \leftarrow s, k \leftarrow k, s+k-t \leftarrow t \), (88) follows from permutation invariance (2) and lastly (89) and (90) follow from friendliness (13) and reducibility (3), respectively. □