

# Wireless Scheduling With Hybrid ARQ

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**Abstract**—A model for downlink wireless scheduling is studied, which takes into account both user-channel conditions and retransmissions with packet combining hybrid [automatic repeat request (ARQ)]. Quality-of-service (QoS) requirements for each user are represented by a cost function, which is an increasing function of queue length. The objective is to find a scheduling rule that minimizes the average cost over time. We consider two scenarios: 1) the cost functions are linear, and packets arrive to the queues according to a Poisson process and 2) the cost functions are increasing, convex, and there are no new arrivals (draining problem). In each case, we transform the system model into a different model that fits into a framework for stochastic scheduling developed by Klimov. Applying Klimov's results, we show that the optimal schedulers for the transformed models in both scenarios are specified by fixed priority rules. Applying the inverse transformation in each case gives the optimal scheduling policy for the original problem. The priorities can be explicitly computed, and in the first scenario, are given by simple closed-form expressions. For the draining problem, we show that the optimal policy never interrupts the retransmissions of a packet. We also show that a simple myopic scheduling policy, called the  $U'R$  rule, performs very close to the optimal scheduling policy in specific cases. We present numerical examples, which compare the performance of the optimal scheduling rule with several heuristic rules.

**Index Terms**—Hybrid automatic repeat request (HARQ), link adaptation, retransmissions, stochastic scheduling, utility function.

## I. INTRODUCTION

SCHEDULING in wireless networks has received considerable attention as a means for providing high-speed data services to mobile users. A basic feature in wireless settings is that the channel quality varies across the user population due to differences in path loss, as well as fading effects. Knowledge of each user's channel quality can be exploited when making scheduling decisions. A variety of such "channel aware" scheduling approaches have been studied recently (e.g., [1]–[6]), and have been incorporated into recent wireless standards.

Manuscript received August 26, 2003; revised April 21, 2004; accepted October 26, 2004. The editor coordinating the review of this paper and approving it for publication is W. Liao. This paper was supported by the Northwestern–Motorola Center for Communications, by the National Science Foundation (NSF) under Grant CCR 9903055, and under CAREER Award CCR-0238382. This paper was presented in part at the 38th Conference on Information Sciences and Systems (CISS'04), Princeton University, Princeton, NJ, March 2004.

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Digital Object Identifier 10.1109/TWC.2005.858025

In this paper, we study scheduling in a wireless network, taking into account packet retransmissions. Link-layer retransmissions are essential for providing reliability over error-prone wireless links. Traditionally, this is accomplished via a standard automatic-repeat-request (ARQ) protocol, where, if a packet cannot be decoded, it is discarded and retransmitted again. Most of the prior work on wireless scheduling either does not consider retransmissions or considers this standard ARQ approach (e.g., [7]). Here, we are interested in hybrid ARQ schemes [8], where the receiver combines all the transmissions of a packet to improve the likelihood of decoding success. A variety of hybrid ARQ techniques have been proposed including diversity combining [9], other "code combining" techniques [10], and incremental redundancy, based on code puncturing [11]. Some recent work in this area includes [12]–[15]. Techniques based on hybrid ARQ are an integral part of many recent wireless standards, such as the Global System for Mobile communications (GSM) Enhanced Data-rate for GSM Evolution (EDGE) system [16]. For our purposes, the key characteristic of these approaches is that each transmission attempt increases the probability of decoding success. The dependence of the probability of decoding success on the number of transmission attempts will vary among the users, depending on their channel conditions. We develop scheduling rules that take this into account.

A goal of any wireless scheduling scheme is to balance the users' quality-of-service (QoS) requirements. Here we represent each user's QoS requirements via a holding cost that is an increasing function of the user's queue length (or equivalently, a "utility" function that is decreasing with the queue length). Our goal is to schedule transmissions to minimize the overall cost. Varying each user's cost function enables the system to tradeoff fairness for throughput, and provides a general framework for scheduling heterogeneous traffic requests. Prior work on scheduling, which assumes a linear or quadratic cost function that depends on the packet delay, is presented in [17]–[20]. Arbitrary increasing delay cost functions have been studied in [19]–[23]. For general cost functions, most authors have focused on developing bounds or heuristic policies. However, in [22] and [23], a generalized  $c\mu$  rule is shown to be optimal for general convex-delay cost in the heavy traffic regime. Here, we consider cost functions that depend on the queue length, instead of the delay of the individual packet. From Little's Law [24], this cost function reflects the average delay of the user's packet with stationary traffic.

We assume a slow-fading environment, which highlights the tradeoff between scheduling efficiency and fairness. This tradeoff can be controlled by the choice of cost function. We remark that our analysis also applies to a fast-fading environment in which the sequence of channel states for each user, indexed by scheduling slots, is independent and identically distributed

(i.i.d.), and scheduler decisions are based on the first-order distribution of channel states (of course, this channel model is only reasonable when the channel appears ergodic over the time scale of scheduling).

Given a set of user cost functions and Poisson packet arrivals, determination of the optimal scheduler with hybrid ARQ can be formulated as a Markov decision process (MDP) and solved via dynamic programming. In general, the solution is complicated and provides little guidance for designing a practical scheduler. To gain insight, we, therefore, consider two special scenarios of practical interest: 1) linear cost functions with Poisson packet arrivals [linear Poisson arrival (LPA) scheduling problem] and 2) general nonlinear increasing convex costs with no new arrivals [draining convex (DC) scheduling problem]. In both cases, our goal is to schedule transmissions to minimize the average cost per packet. Linear cost functions can take into account relative priorities by assigning different weights to the different queues.<sup>1</sup> (If the weights are the same, then the performance metric becomes total throughput.) Nonlinear cost functions include linear cost functions with a cost that is independent of the number of retransmissions as a special case, and can capture different types of delay requirements such as deadlines.

We show that both scheduling problems can be transformed into special cases of a classic scheduling problem solved by Klimov [25]. As a consequence, the optimal schedulers for the transformed system in both scenarios are specified by fixed priority rules. We can then map the priority rules back to get the optimal scheduling policy for the original problem. These priorities can be explicitly computed, and for the LPA problem, are given by simple closed-form expressions. Casting the DC problem into the Klimov framework requires a more complicated transformation. Applying Klimov's results gives an iterative algorithm for computing the priority indices. We show that the optimal policy never interrupts retransmissions of a packet in order to transmit another packet. We also formulate the DC problem as an MDP, and show that the priority increases with queue length.

Finally, we consider a simple myopic scheduling policy called the  $U'R$  rule, which takes both the channel condition and cost function into consideration [26]. We give scenarios for which the  $U'R$  rule performs close to the optimal policy. We also give a numerical comparison of the performance of the optimal rule with other heuristic policies.

The rest of the paper is organized as follows. In Section II, we describe the system model, and we briefly review the Klimov scheduling model [25] in Section III. Solutions to the LPA and DC problems are presented in Sections IV and V, respectively. Section VI presents some numerical examples, and the conclusion is presented in Section VII.

## II. SYSTEM MODEL

We consider a set of  $N$  mobile users served by a single base station or access point. Our focus is on downlink traffic (from

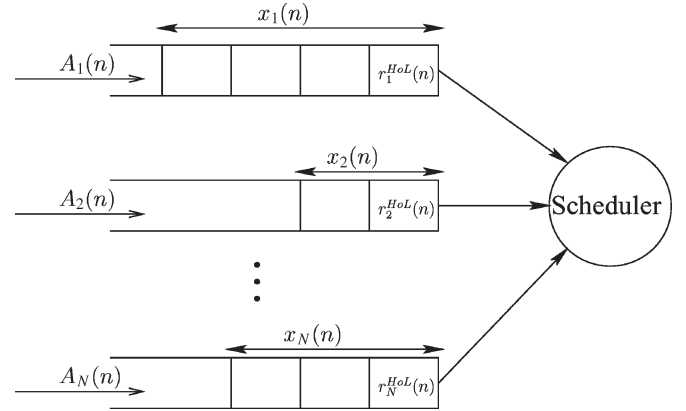


Fig. 1. System model.

the base station to the users). As shown in Fig. 1, packets for each user arrive at the base station according to independent random processes, and are accumulated in  $N$  queues until they are served. The base station transmits to one user at a time in slots of fixed durations. In each slot, a scheduler decides which packet to transmit. We assume that the scheduler is restricted to choosing an HoL packet from one of the queues.

When the receiver is unable to decode a transmission successfully, the packet stays at the HoL, and is retransmitted until it is decoded successfully. We ignore feedback delay, i.e., the packet becomes immediately available for retransmission after decoding failure [14]. This approximates the case when the feedback delay is small compared to the transmission time of a packet. Given that a packet for user  $i$  has not been successfully decoded in  $r_i$  transmission attempts, let  $g_i(r_i)$  denote the probability of decoding failure for the next transmission. This depends on the specific hybrid ARQ scheme, and on user  $i$ 's channel conditions. We assume that  $g_i(\cdot)$  is time invariant. This is reasonable in a slow-fading environment, where each user's channel is constant over the time scale of interest. An empirical method to estimate  $g_i(\cdot)$  in this case is presented in [12]. We remark that a time-invariant  $g_i(\cdot)$  also applies to the fast-fading situation in which the sequence of channel states over slots for each user is i.i.d. In that case,  $g_i(\cdot)$  is averaged over the first-order channel distribution.

We also assume that  $g_i(\cdot)$  is nonincreasing in the number of transmission attempts  $r_i$ , i.e.,  $g_i(r_i) \geq g_i(r'_i)$  for all  $r_i \leq r'_i$ . This will be satisfied by any reasonable hybrid ARQ approach. To simplify our analysis, we assume that there is a maximum number of transmission attempts  $r_i^{\text{max}}$  for user  $i$ , and that  $g_i(r_i^{\text{max}}) = 0$ , i.e., a packet is always successfully decoded after  $r_i^{\text{max}} + 1$  transmissions.<sup>2</sup> Note that the special case  $g_i(r_i) = g_i(0)$  for all  $r_i$  models standard ARQ.

Let  $A_i(n)$  denote the arrivals for user  $i$  during the  $n$ th slot, which is independent of the arrivals for the other users. Let  $\mathcal{S}(n) = (r_1^{\text{HoL}}(n), \dots, r_N^{\text{HoL}}(n); x_1(n), \dots, x_N(n))$  be the state vector at the  $n$ th decision epoch (i.e., the start of the  $n$ th time slot), where  $r_i^{\text{HoL}}(n) \in \{0, 1, \dots, r_i^{\text{max}}\}$  is the number of transmission attempts for the  $i$ th HoL packet, and  $x_i(n) \in \{0, 1, \dots\}$  is the queue length for user  $i$ .

<sup>1</sup>In our case, linear cost implies a weighted combination of queue length and number of retransmissions for the head-of-line (HoL) packet.

<sup>2</sup>Equivalently, after the last transmission attempt the packet is dropped; the cost function can be modified to reflect a penalty for this occurrence.

Given  $\mathbf{S}(n)$ , the scheduler must determine which HoL packet should be transmitted in the  $n$ th slot. A scheduling policy  $\pi$  is defined to be a mapping from each state vector to an index in  $\{0, 1, \dots, N\}$ . If  $\pi(\mathbf{S}(n)) = i$ , the scheduler transmits the HoL packet of queue  $i$ ; if  $\pi(\mathbf{S}(n)) = 0$ , the scheduler idles and no packet is transmitted. Given a policy  $\pi$ , the state  $\mathbf{S}(n)$  evolves according to

$$r_i^{\text{HoL}}(n+1) = \begin{cases} 0, & \pi(\mathbf{S}(n)) = i \text{ and success} \\ r_i^{\text{HoL}}(n) + 1, & \pi(\mathbf{S}(n)) = i \text{ and failure} \\ r_i^{\text{HoL}}(n), & \pi(\mathbf{S}(n)) \neq i \end{cases} \quad (1)$$

and

$$x_i(n+1) = \begin{cases} x_i(n) + A_i(n) - 1, & \pi(\mathbf{S}(n)) = i \text{ and success} \\ x_i(n) + A_i(n), & \text{otherwise.} \end{cases} \quad (2)$$

Here ‘‘success’’ and ‘‘failure’’ refer to the decoding outcome for the given transmission. We restrict ourselves to the set of feasible policies  $\Pi$ , which contains all nonidling, nonpreemptive, nonanticipative, and stationary policies.<sup>3</sup>

Each user  $i$  has cost function  $U_i(x_i(n), r_i^{\text{HoL}}(n))$  associated with the start of the  $n$ th slot. This cost function is increasing and convex in  $x_i(n)$ , i.e., for  $x_1 > x_2$ ,  $U_i(x_1, y) > U_i(x_2, y)$ , and  $(\partial U_i(x, y)/\partial x)|_{x=x_1} > (\partial U_i(x, y)/\partial x)|_{x=x_2}$  (assuming  $U_i(x_i(n), r_i^{\text{HoL}}(n))$  is differentiable at  $x_i(n) = x_1$  and  $x_i(n) = x_2$ ). We assume that  $U_i(0, 0) = 0$ , i.e., there is no holding cost for an empty queue. Different cost functions reflect different QoS requirements or priorities for the users.

We consider two scenarios. In the LPA problem, to be discussed in Section IV, packets arrive to the queues according to independent Poisson processes with rates  $\lambda_i$ ,  $i = 1, \dots, N$ . The cost function for user  $i$  is linear,<sup>4</sup> and is given by

$$U_i(x_i(n), r_i^{\text{HoL}}(n)) = \begin{cases} c_{i,0}(x_i(n) - 1) + c_{i,r_i^{\text{HoL}}(n)}, & x_i(n) > 0 \\ 0, & x_i(n) = 0 \end{cases} \quad (3)$$

where  $c_{i,r_i}$  is the holding-cost rate (cost per unit time per packet) for a packet of user  $i$  with  $r_i$  transmission attempts. For all  $i$

$$0 \leq c_{i,r_i} \leq c_{i,r'_i}, \quad r_i < r'_i \quad (4)$$

which means that the greater the number of transmission attempts, the higher the holding cost. The LPA problem is to find  $\pi \in \Pi$  that minimizes the long-term average expected cost

$$J_{\text{LPA}} = \lim_{\tau \rightarrow \infty} \frac{1}{\tau} E_{\pi} \left[ \sum_{n=1}^{\tau} \sum_{i=1}^N U_i(x_i(n), r_i^{\text{HoL}}(n)) \right]. \quad (5)$$

<sup>3</sup>A policy is nonpreemptive if the transmission of a packet is not interrupted by an arrival, and is nonanticipative if it does not account for future decoding results or arrivals.

<sup>4</sup>More precisely, this is an affine cost function, because of the additive constant associated with the HoL packet.

In the second scenario, discussed in Section V, we consider a draining problem with no new arrivals (i.e.,  $A_i(n) = 0$  for all  $i$  and  $n$ ). In this case, we allow the cost to be an arbitrary increasing convex function of the queue length, and independent of the number of transmission attempts, i.e.,  $U_i(x_i(n), r_i^{\text{HoL}}(n)) = U_i(x_i(n))$ . We refer to this as the DC problem. Given an initial batch of packets  $(x_1(1), \dots, x_N(1))$ , the goal is to find  $\pi \in \Pi$ , which minimizes the total expected draining cost, i.e.,

$$J_{\text{DC}} = E_{\pi} \left[ \sum_{n=1}^{\infty} \sum_{i=1}^N U_i(x_i(n)) \right]. \quad (6)$$

This can also be interpreted as a model for a system with correlated batch arrivals [27], where the interarrival time is long enough to finish each batch before the arrival of a new batch (an example application is simultaneous downloads to multiple users).

The need to track the number of transmission attempts of every HoL packet complicates the scheduling. Our analysis is based on the Klimov model [25], which is described next.

### III. KLIMOV MODEL

The Klimov model [25] has a single nonpreemptive server, which is allocated to the jobs in a network of  $K$   $M/G/1$  queues. Jobs arrive according to a Poisson process with rate  $\lambda$ , and are assigned to queue  $m$  with probability  $p_m$ , where  $\sum_{m=1}^K p_m = 1$ . The service time for a job at queue  $m$  ( $m = 1, \dots, K$ ) has distribution function  $B_m(x)$ , and finite mean  $b_m$ . After service completion at queue  $m$ , a job enters queue  $j$  ( $j = 1, \dots, K$ ) with probability  $p_{mj}$ , or leaves the system with the probability  $1 - \sum_{j=1}^K p_{mj}$ . The transition matrix  $P = [p_{mj}, 1 \leq m, j \leq K]$  is such that every job eventually leaves the system, i.e.,  $\lim_{n \rightarrow \infty} P^n = 0$ . The arrival rate is assumed to not exceed the processing capacity of the system, i.e.,  $\lambda \mathbf{p}(I - P)^{-1} \mathbf{b} < 1$ , where  $\mathbf{p} = (p_1, \dots, p_K)$  and  $\mathbf{b} = (b_1, \dots, b_K)$ .<sup>5</sup>

The objective is to find a feasible scheduling policy  $\pi$  that minimizes a linear combination of the time-averaged number of jobs at each queue,

$$J_{\text{KM}} = \lim_{\tau \rightarrow \infty} \frac{1}{\tau} E_{\pi} \left[ \int_0^{\tau} \sum_{m=1}^K c_m x_m(t) dt \right] \quad (7)$$

where  $x_m(t)$  is the number of jobs in queue  $m$  at time  $t$  and  $c_m \geq 0$  is a (linear) holding-cost rate for queue  $m$ .

The optimal scheduling policy is a fixed-priority rule [25]. This means that a time-invariant priority index can be calculated for each queue, which is independent of the arrival process and queue lengths. At each decision epoch, the server serves a job from the nonempty queue with the highest priority.

The optimal priority indices can be calculated via an iterative algorithm [25], which starts from the set of queues  $\Omega = \{1, 2, \dots, K\}$  and selects the lowest priority queue at each iteration. Given a subset of queues  $M \subset \Omega$ , the priority for queue  $m \in M$  is determined by  $C_m^{(M)}/T_m^{(M)}$ , where  $C_m^{(M)}$  is

<sup>5</sup> $(\mathbf{x})'$  denotes the transpose of  $\mathbf{x}$ .

the equivalent holding-cost rate, and  $T_m^{(M)}$  is the average total service time (not including waiting time) for a job in queue  $m$  (i.e., until it exits from  $M$ ). Since the service times are independent, for each  $m \in M$

$$T_m^{(M)} = \sum_{j \in M} p_{mj} T_j^{(M)} + b_m. \quad (8)$$

The optimal priority indices are computed by the following Klimov algorithm:

- 1) Initialization:  $M_K = \Omega$ ,  $C_m^{(M_K)} = c_m$  for all  $m \in M_K$ ,  $k = K$ ;
- 2) Find a queue  $\alpha_k$  with lowest priority, i.e.,

$$\alpha_k = \arg \min_{m \in M_k} \left\{ \frac{C_m^{(M_k)}}{T_m^{(M_k)}} \right\} \quad (9)$$

with ties broken arbitrarily.

- 3)  $M_{k-1} = M_k - \{\alpha_k\}$
- 4) If  $M_{k-1} = \phi$  (null set), then stop. Otherwise, for each  $m \in M_{k-1}$ , compute

$$C_m^{(M_{k-1})} = T_m^{(M_k)} \left[ \frac{C_m^{(M_k)}}{T_m^{(M_k)}} - \frac{C_{\alpha_k}^{(M_k)}}{T_{\alpha_k}^{(M_k)}} \right]. \quad (10)$$

Decrement  $k$  and go to step 2.

In this way, the queues are ordered in descending priorities,  $\alpha_1 \geq \alpha_2 \geq \dots \geq \alpha_K$ , where  $(\alpha_1, \alpha_2, \dots, \alpha_K)$  is a permutation of queue indices  $(1, 2, \dots, K)$ . The optimal policy  $\pi$  always assigns the server to the nonempty queue  $\alpha_k$  with the smallest index  $k$ . Moreover, this scheduler minimizes the total holding cost for each busy period of the system, starting from any initial state [28].

When discussing the DC problem, we consider a variation of the Klimov model, which we call the draining Klimov model. In this case, the goal is to find a policy, which minimizes the total expected holding cost for a batch of packets initially in the system with no new arrivals. The priority rule specified by the Klimov algorithm is also optimal for the draining model, since the scheduler minimizes the total holding cost of each busy period. In other words, the draining problem can be viewed as a special busy period with no further arrivals.

#### IV. LPA SCHEDULING PROBLEM

In this section, we reformulate the LPA scheduling problem as a special case of Klimov's problem, which we refer to as the LPAK scheduling problem. We will show that the optimal scheduling policy for the LPAK problem is also optimal for the LPA problem.

##### A. LPAK Scheduling Problem

The LPAK problem is a relaxation of LPA with respect to the service discipline. In LPA, there is one queue for each user  $i$ , and the HoL packet in a queue has priority over all the other packets in the queue. The LPAK problem is illustrated in Fig. 2 for two users with  $r_1^{\max} = 2$  and  $r_2^{\max} = 1$ . There are  $r_i^{\max} + 1$  queues for each user  $i$ , and each queue is labeled by  $(i, r_i)$

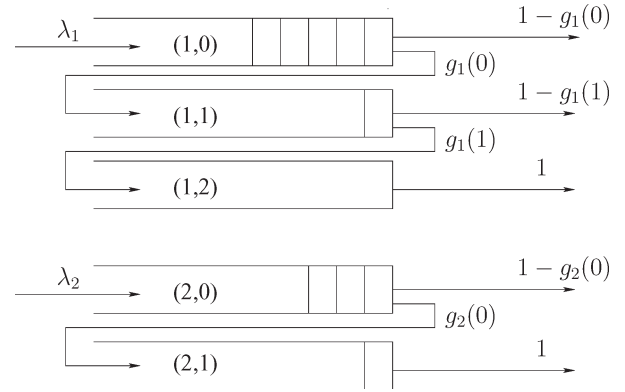


Fig. 2. LPAK-system model.

for  $r_i = 0, \dots, r_i^{\max}$ , where  $r_i$  is the number of transmission attempts for all packets in the queue. There are a total of  $K = \sum_{i=1}^N (r_i^{\max} + 1)$  queues. For the example in Fig. 2,  $K = 3 + 2 = 5$ . At each decision epoch, the server decides which of the  $K$  HoL packets to serve. Because of the additional queues in the LPAK problem, the HoL packet corresponding to a particular user (in the original LPA problem) does not necessarily have priority over the user's other packets. This relaxation makes LPAK a standard Klimov problem. Subsequently, we will show that the optimal scheduling rule for LPAK still gives priority to the user's HoL packet.

The arrival process is Poisson with rate  $\lambda = \sum_{i=1}^N \lambda_i$ , and each packet is assigned to queue  $(i, 0)$  with probability  $p_{i,0} = \lambda_i / \lambda$ . The service time for each queue  $(i, r_i)$  is deterministic with  $b_{i,r_i} = 1$  time slot. The transition probabilities among queues are determined by the probability of decoding failure. That is, after a packet from queue  $(i, r_i)$ ,  $r_i < r_i^{\max}$  has been served, it enters queue  $(i, r_i + 1)$  with probability  $p_{(i,r_i),(i,r_i+1)} = g_i(r_i)$ , corresponding to a decoding failure, or leaves the system with probability  $1 - g_i(r_i)$ , corresponding to a decoding success. After a packet from queue  $(i, r_i^{\max})$  has been served, it leaves the system with probability 1. Thus

$$p_{(i,r_i),(j,r_j)} = \begin{cases} g_i(r_i), & r_i < r_i^{\max}, (j, r_j) = (i, r_i + 1) \\ 0, & \text{otherwise.} \end{cases} \quad (11)$$

For any set  $M \subset \Omega = \{1, \dots, K\}$  and  $(i, r_i) \in M$ , the average total service time is

$$T_{i,r_i}^{(M)} = \sum_{(j,r_j) \in M} p_{(i,r_i),(j,r_j)} T_{j,r_j}^{(M)} + 1. \quad (12)$$

The holding-cost rate of queue  $(i, r_i)$  is  $c_{i,r_i}$ , and the number of packets in queue  $(i, r_i)$  at the  $n$ th decision epoch is  $x_{i,r_i}(n)$ . The goal is to find a policy  $\pi \in \Pi$ , which minimizes the time-averaged expected cost

$$J_{\text{LPAK}} = \lim_{\tau \rightarrow \infty} \frac{1}{\tau} E_{\pi} \left[ \sum_{n=1}^{\tau} \sum_{(i,r_i) \in \Omega} c_{i,r_i} x_{i,r_i}(n) \right]. \quad (13)$$

**B. Optimal Policies for LPAK and LPA Scheduling Problems**

For the LPAK scheduling problem, the optimal priority indices can be calculated iteratively using the Klimov algorithm in Section II. Consider this algorithm with the following rule used to break any ties that occur in (9): when a tie occurs, set  $\alpha_k$  to be the queue  $(i, r_i)$  such that for all other queues  $(j, r_j)$  in the tie,  $j > i$ , or  $j = i$ , and  $r_j > r_i$ .

*Lemma 1:* Let  $M_k, k = 1, 2, \dots, K$ , be the sets generated by applying the Klimov algorithm to the LPAK problem with the preceding tie-breaking rule. For each  $k = 1, \dots, K$  and for all  $(i, r_i) \in M_k$ , the following properties hold:

- 1)  $(i, r'_i) \in M_k$ , for all  $r'_i > r_i$ .
- 2)  $T_{i,r_i}^{(M_k)} = 1 + \sum_{j=r_i}^{r_i^{\max}-1} \prod_{l=r_i}^j g_i(l) = T_{i,r_i}^{(\Omega)}$ .
- 3)  $T_{i,r'_i}^{(M_k)} \leq T_{i,r_i}^{(M_k)}$ , for all  $r'_i > r_i$ .
- 4)  $\alpha_k = \arg \min_{(i,r_i) \in M_k} (c_{i,r_i}/T_{i,r_i}^{(\Omega)})$ .

Property 1) shows that each set  $M_k$  is a ‘‘threshold set,’’ i.e., for each user  $i$ , there is a threshold  $r_i^*$ , such that  $(i, r'_i) \in M_k$  if and only if  $r'_i \geq r_i^*$ . Property 2) shows that  $T_{i,r_i}^{(\Omega)}$  only depends on  $g_i(r'_i)$  for  $r'_i \geq r_i$ , and therefore, only depends on the presence of queues  $(i, r'_i)$ ,  $r'_i \geq r_i$ , in the set  $M_k$ . Thus,  $T_{i,r_i}^{(M_k)} = T_{i,r_i}^{(\Omega)}$  for every  $k$  and every  $(i, r_i) \in M_k$ , i.e., the service times are fixed for each iteration. Property 3) states that the service times  $T_{i,r_i}^{(M_k)}$  are nonincreasing in  $r_i$ . This follows directly from 2). Property 4) states that the optimal priority order can be calculated directly without any iterations. From 2), the equivalent holding cost in (10) can be written as

$$C_{i,r_i}^{(M_k)} = c_{i,r_i} - T_{i,r_i}^{(\Omega)} \sum_{l=k+1}^K \frac{C_{\alpha_l}^{(M_l)}}{T_{\alpha_l}^{(M_l)}}. \quad (14)$$

From this, we have that

$$\frac{C_{i,r_i}^{(M_k)}}{T_{i,r_i}^{(M_k)}} = \frac{c_{i,r_i}}{T_{i,r_i}^{(\Omega)}} - \sum_{l=k+1}^K \frac{C_{\alpha_l}^{(M_l)}}{T_{\alpha_l}^{(M_l)}} \quad (15)$$

and 4) follows. A detailed proof is omitted.

*Theorem 1:* For the LPAK scheduling problem, the optimal scheduling policy is a fixed priority rule in which the priorities,  $\alpha_1, \alpha_2, \dots, \alpha_K$ , satisfy

$$\frac{C_{\alpha_1}}{T_{\alpha_1}^{(\Omega)}} \geq \frac{C_{\alpha_2}}{T_{\alpha_2}^{(\Omega)}} \geq \dots \geq \frac{C_{\alpha_K}}{T_{\alpha_K}^{(\Omega)}}. \quad (16)$$

This follows from the main theorem in [25] and Lemma 1.

To derive the optimal LPA scheduler, let  $\mathbf{R} = (r_1^{\text{HoL}}, r_2^{\text{HoL}}, \dots, r_N^{\text{HoL}})$  denote the vector of retransmission attempts for HoL packets across the  $N$  queues. Let  $T_{i,r_i^{\text{HoL}}}$  be the expected total service time for user  $i$ 's HoL packet (not including any waiting time) until it exits the system, which is given by

$$T_{i,r_i^{\text{HoL}}} = 1 + \sum_{j=r_i^{\text{HoL}}}^{r_i^{\max}-1} \prod_{l=r_i^{\text{HoL}}}^j g_i(l). \quad (17)$$

*Corollary 1:* For the LPA scheduling problem, the optimal scheduling rule is to transmit the HoL packet with the highest priority index  $c_{i,r_i^{\text{HoL}}}/T_{i,r_i^{\text{HoL}}}$  among all nonempty queues.

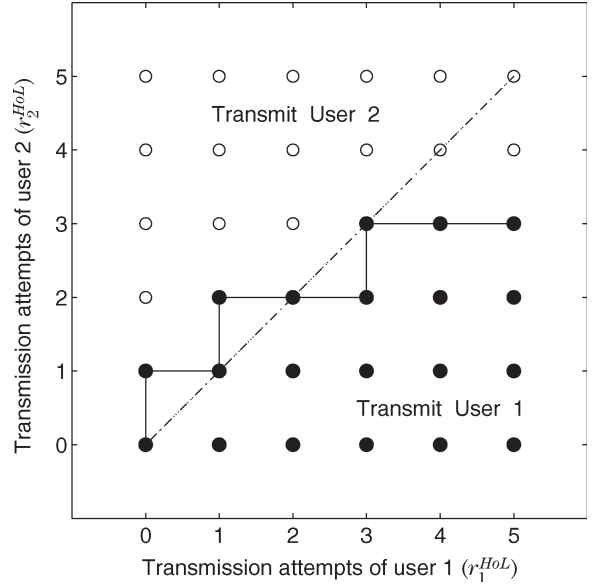


Fig. 3. Optimal scheduling policy as a function of the transmission attempts for two users in the LPA problem. A dot (circle) means it is optimal to transmit the HoL packet for user 1 (user 2).

Furthermore, the optimal policy is a monotonic threshold policy on the number of transmission attempts, i.e., if it is optimal to transmit user  $i$  when  $\mathbf{R} = (r_1^{\text{HoL}}, \dots, r_i^{\text{HoL}}, \dots, r_N^{\text{HoL}})$ , then it is optimal to transmit user  $i$  when  $r_i^{\text{HoL}}$  is replaced by  $r_i^{\text{HoL}} > r_i^{\text{HoL}}$ .

*Proof:* See the Appendix. ■

The optimal LPA scheduling rule depends on the set of holding-cost rates  $c_{i,r_i}$ , the number of transmission attempts  $r_i^{\text{HoL}}$ , and the probability of decoding success  $g_i(\cdot)$  (i.e., the channel condition) across all users. A higher cost rate, more transmission attempts, or a better channel results in a higher priority. Notice that scheduling decisions do not explicitly depend on the arrival processes or queue lengths, although the latter affect the holding costs.

Computing the priority indices via the Klimov algorithm generally requires  $K$  iterations with computational complexity of  $O(K^2)$ . For the LPA problem, due to the special structure of the transition matrix and the deterministic service times, we obtain simple closed-form formulas for the priority indices with associated complexity  $O(K)$ . This may be suitable for online scheduling with time-varying channel conditions.

We illustrate the optimal scheduling policy with some numerical examples. Consider a system with  $N = 2$  users, and probability of decoding failure

$$g_i(r_i) = \begin{cases} \eta_i \cdot 0.5^{r_i}, & 0 \leq r_i < r_i^{\max} \\ 0, & r_i = r_i^{\max} \end{cases} \quad (18)$$

for  $i = 1, 2$ . That is, the initial probability of decoding failure is  $\eta_i$ , and is reduced by half with each retransmission until  $r_i = r_i^{\max}$ . This type of exponentially decreasing  $g_i(r_i)$  is motivated by numerical results in [12].

Fig. 3 shows the optimal scheduling policy as a function of the number of transmission attempts for each user. The parameters are  $(\eta_1, r_1^{\max}) = (0.02, 5)$ ,  $(\eta_2, r_2^{\max}) = (0.1, 5)$ ,  $c_{1,r_1} = 1$



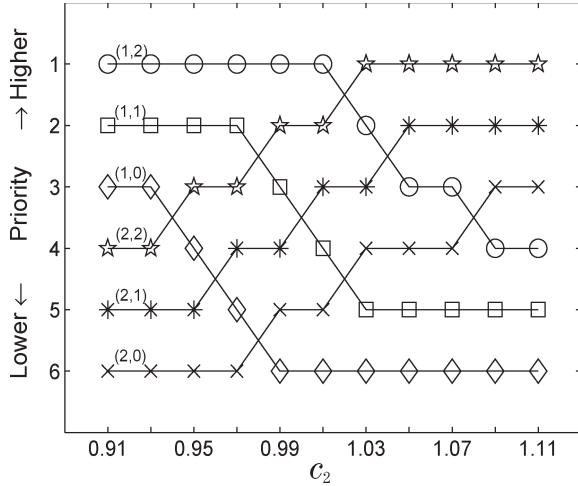


Fig. 4. Optimal priority orders versus holding-cost rate of user 2 in the LPAK problem.

(for all  $r_1$ ), and  $c_{2,r_2} = 1.01$  (for all  $r_2$ ). In this case, user 1 has the better channel, but has a slightly lower holding cost than user 2. As stated in Corollary 1, the optimal scheduling policy is a monotonic threshold policy on  $r_i^{\text{HoL}}(i = 1, 2)$ ; the threshold is shown by the solid line in Fig. 3. Comparing this with the dash-dotted line  $r_1^{\text{HoL}} = r_2^{\text{HoL}} = r$ , when  $r$  is small ( $r \leq 3$ ), user 1 has priority because of the better channel (smaller  $T_{i,r}$ ). However, when  $r$  is large ( $r > 3$ ), user 2 has priority. The reason is that  $g_i(r)$  is very small, which makes  $T_{i,r}$  very close to 1 for both users. Thus, the difference between the cost rates  $c_{i,r}$  is the main factor in determining the priority order.

Fig. 4 shows the optimal priority orders versus the holding-cost rate of user 2. In this case, both users have the same channel conditions  $(\eta_1, r_1^{\text{max}}) = (\eta_2, r_2^{\text{max}}) = (0.05, 2)$ . There are six types of packets in the system,  $(i, r_i)$ ,  $i = 1, 2$ ,  $r_i = 0, 1, 2$ , and their priorities are ordered from 1 (highest) to 6 (lowest). The holding-cost rates for user 1 are  $c_{1,0} = 0.98$ ,  $c_{1,1} = 1$ , and  $c_{1,2} = 1.02$ . The holding-cost rates for user 2 are  $c_{2,0} = c_{2,1} = c_{2,2} \triangleq c_2$ , which varies from 0.91 to 1.11. Fig. 4 shows that the packet priorities increase with  $r_i$ . This reflects the fact that the HoL packet has priority over the other user's packets. At  $c_2 = 0.91$ , user 1 has priority over user 2. Hence, a new packet arrival for user 1 has priority over a retransmission from user 2. Of course, as  $c_2$  increases, the priorities for user 2 increase from lowest (4,5,6) to highest (1,2,3).

## V. DC SCHEDULING PROBLEM

For the DC problem, the cost function can be nonlinear, which precludes a direct association with the Klimov model. We circumvent this difficulty by again transforming the problem into a related Klimov problem with more queues. We refer to the latter problem as the DC Klimov (DCK) scheduling problem. Applying the Klimov algorithm, we show that it is not optimal to interrupt the retransmission of a packet. We then formulate the DC problem with two users as an MDP, and show that the optimal scheduling rule is a monotonic threshold policy on the queue lengths.

### A. DCK Scheduling Problem

We construct a mapping between the DC and DCK models. Let  $A_i$  be the number of user  $i$ 's packets initially in the system in the DC model. Each queue in the DC model is replaced by  $K_i = (r_i^{\text{max}} + 1)A_i$  queues in the DCK model. Assume, for the DC model, that at time  $n$ , user  $i$ 's queue length is  $x_i(n) > 0$ , with holding cost  $U_i(x_i(n))$ , and the number of transmission attempts of the HoL packet is  $r_i^{\text{HoL}}(n)$ . In the DCK model, this corresponds to there being one packet in the queue  $(i, r_i^{\text{HoL}}(n), x_i(n))$  with linear holding-cost rate  $c_{i,r_i^{\text{HoL}}(n), x_i(n)} = U_i(x_i(n))$ , and no packets in any of the other  $K_i - 1$  queues corresponding to user  $i$ .

Let  $\Omega$  denote the set of all  $K = \sum_{i=1}^N K_i$  queues in the DCK model. The service time for each queue  $(i, r_i, x_i) \in \Omega$  is still deterministic with  $b_{i,r_i,x_i} = 1$ . Suppose that user  $i$ 's HoL packet is transmitted during slot  $n$  (DC model). Then, in the DCK model, the corresponding packet in queue  $(i, r_i^{\text{HoL}}(n), x_i(n))$  either: 1) enters queue  $(i, r_i^{\text{HoL}}(n) + 1, x_i(n))$  with probability  $g_i(r_i^{\text{HoL}}(n))$  (decoding fails); 2) enters queue  $(i, 0, x_i(n) - 1)$  with probability  $1 - g_i(r_i^{\text{HoL}}(n))$  (decoding succeeds and  $x_i(n) > 1$ ), or 3) leaves the system ( $r_i = r_{\text{max}}$ ). The transition probabilities in the DCK model are therefore given by

$$P^{(i,r_i,x_i),(j,r_j,x_j)} = \begin{cases} g_i(r_i), & r_i < r_i^{\text{max}}, \quad (j, r_j, x_j) = (i, r_i + 1, x_i) \\ 1 - g_i(r_i), & x_i > 1, \quad (j, r_j, x_j) = (i, 0, x_i - 1) \\ 0, & \text{otherwise.} \end{cases} \quad (19)$$

For any set  $M \subset \Omega$  and queue  $(i, r_i, x_i) \in M$ , the average total service time is

$$T_{i,r_i,x_i}^{(M)} = \sum_{(j,r_j,x_j) \in M} P^{(i,r_i,x_i),(j,r_j,x_j)} T_{j,r_j,x_j}^{(M)} + 1. \quad (20)$$

The DCK scheduling problem is to find a scheduling policy  $\pi \in \Pi$  that minimizes the total expected holding cost for draining all the packets, i.e.,

$$J_{\text{DCK}} = E_{\pi} \left[ \sum_{n=1}^{\infty} \sum_{(i,r_i,x_i) \in \Omega} \mathbf{1}_{i,r_i,x_i}(n) U_i(x_i) \right] \quad (21)$$

where

$$\mathbf{1}_{i,r_i,x_i}(n) = \begin{cases} 1, & (i, r_i, x_i) \text{ is nonempty in slot } n \\ 0, & (i, r_i, x_i) \text{ is empty in slot } n. \end{cases} \quad (22)$$

The DCK problem is therefore a special case of Klimov's scheduling problem. Hence, we can apply the Klimov algorithm to calculate the optimal priorities, which in turn solves the DC problem.

Unlike the LPA problem, for the DC problem, the priorities cannot be computed in closed form. However, we can characterize some basic properties of the optimal policy.

### B. Properties of the Optimal Scheduler

As in Section IV-B, consider the Klimov algorithm with the following rule to break any tie that occurs in (9): set  $\alpha_k = (i, r_i, x_i)$  so that for all other queues  $(j, r_j, x_j)$  in the tie,  $j > i$ .

*Lemma 2:* Let  $M_k, k = 1, \dots, K$ , be the sets generated by applying the Klimov algorithm to the DCK problem with the preceding tie-breaking rule. For each  $k$ , for all  $(i, r_i, x_i) \in M_k$ , and for all  $r'_i > r_i$ , the following properties hold:

- 1)  $(i, r'_i, x_i) \in M_k$ ;
- 2)  $T_{i, r'_i, x_i}^{(M_k)} \leq T_{i, r_i, x_i}^{(M_k)}$ ;
- 3)  $C_{i, r'_i, x_i}^{(M_k)} \geq C_{i, r_i, x_i}^{(M_k)}$ .

Property 1) shows that each set  $M_k$  is a ‘‘threshold set,’’ i.e., for each user  $i$  and queue length  $x_i$ , there is a threshold  $r_i^*$ , such that  $(i, r'_i, x_i) \in M_k$  if and only if  $r'_i \geq r_i^*$ . Although there is no direct relationship between  $T_{i, r_i, x_i}^{(M_k)}$  and  $T_{i, r_i, x_i}^{(\Omega)}$ , as in the LPAK problem, 2) states that  $T_{i, r_i, x_i}^{(M_k)}$  is still nonincreasing in  $r_i$ . Property 3) states that the equivalent holding-cost rate  $C_{i, r_i, x_i}^{(M_k)}$  is nondecreasing in  $r_i$ . This follows from the monotonicity and convexity of  $U_i(\cdot)$ . A detailed proof is omitted.

*Theorem 2:* The optimal DCK scheduler assigns queue  $(i, r'_i, x_i)$  higher priority than queue  $(i, r_i, x_i)$  for all  $i, x_i$ , and  $r'_i > r_i$ .

*Proof:* Suppose queue  $(i, r_i, x_i)$  has a higher priority than that for queue  $(i, r'_i, x_i)$ , where  $r'_i > r_i$ . Then there exists a  $k$  such that  $(i, r_i, x_i) \in M_k$ , but  $(i, r'_i, x_i) \notin M_k$ , which contradicts property 1) of Lemma 2. ■

*Corollary 2:* Once the optimal DC scheduler starts to transmit a packet to user  $i$ , it continues to transmit the packet until it is successfully decoded.

*Proof:* Assume that at time  $n$ , the DC scheduler transmits a new packet to user  $i$  with queue length  $x_i(n)$ . In the DCK problem, this corresponds to queue  $(i, 0, x_i(n))$  having the highest priority among all nonempty queues. If decoding fails, the packet leaves  $(i, 0, x_i(n))$  and enters  $(i, 1, x_i(n))$  at time  $(n + 1)$ . According to Theorem 2,  $(i, 1, x_i(n))$  has higher priority than  $(i, 0, x_i(n))$ . Since the priorities of all other packets in the DCK problem remain unchanged,  $(i, 1, x_i(n))$  must have the highest priority at time  $(n + 1)$ . Iterating this argument, user  $i$  has the highest priority until the corresponding DCK ‘‘packet’’ enters  $(i, 0, x_i(n) - 1)$  (or if  $x_i(n) = 1$ , the packet leaves the system). This corresponds to transmitting the HoL packet for user  $i$  until it is successfully decoded. ■

Note that Corollary 2 is not true for the LPA problem, as shown in Fig. 4. The key difference here is that there are no arrivals that can change the priority orders among the users during a retransmission. Another difference is that the DC optimal scheduler depends on the queue lengths in a complicated way, which depends on the specific choice of cost function.

### C. Markov Decision Formulation

In this section, we formulate the DC problem as an MDP. To simplify the discussion, we consider only two users. The system state space is  $S = \{(r_1, r_2, x_1, x_2) | 0 \leq r_i \leq r_i^{\max}, 0 \leq x_i \leq A_i, i \in \{1, 2\}\}$ . The action space is  $V = \{v_0, v_1, v_2\}$ , where  $v_0$

represents idling (if there is no packet in the system), and  $v_i$  represents transmitting the HoL packet of user  $i, i = 1, 2$ .

The scheduling problem can be formulated as a stochastic shortest path problem over an infinite time horizon [29]. Let  $J(r_1, r_2, x_1, x_2)$  denote the optimal cost-to-go starting from state  $(r_1, r_2, x_1, x_2)$ . This must satisfy Bellman’s equation [29], which gives the following conditions:

- 1) if  $x_1 = x_2 = 0$ , then  $J(0, 0, 0, 0) = 0$ ;
- 2) If  $x_1 > 0$  and  $x_2 = 0$ , then

$$J(r_1, 0, x_1, 0) = U_1(x_1) + [1 - g_1(r_1)] J(0, 0, x_1 - 1, 0) + g_1(r_1) J(\min(r_1 + 1, r_1^{\max}), 0, x_1, 0)$$

- 3) If  $x_1 = 0$  and  $x_2 > 0$ , then

$$J(0, r_2, 0, x_2) = U_2(x_2) + [1 - g_2(r_2)] J(0, 0, 0, x_2 - 1) + g_2(r_2) J(0, \min(r_2 + 1, r_2^{\max}), 0, x_2).$$

- 4) If  $x_1 > 0$  and  $x_2 > 0$ , then

$$\begin{aligned} J(r_1, r_2, x_1, x_2) &= U_1(x_1) + U_2(x_2) \\ &+ \min \{ [1 - g_1(r_1)] J(0, r_2, x_1 - 1, x_2) \\ &+ g_1(r_1) J(\min(r_1 + 1, r_1^{\max}), r_2, x_1, x_2), \\ &[1 - g_2(r_2)] J(r_1, 0, x_1, x_2 - 1) \\ &+ g_2(r_2) J(r_1, \min(r_2 + 1, r_2^{\max}), x_1, x_2) \}. \end{aligned}$$

Note that it is never possible for  $x_i = 0$  and  $r_i > 0$ .

*Lemma 3:* The optimal cost-to-go has the following property

$$\begin{aligned} &[1 - g_2(r_2)] J(r_1, 0, x_1, x_2 - 1) \\ &+ g_2(r_2) J(r_1, \min(r_2 + 1, r_2^{\max}), x_1, x_2) \\ &- [1 - g_1(r_1)] J(0, r_2, x_1 - 1, x_2) \\ &- g_1(r_1) J(\min(r_1 + 1, r_1^{\max}), r_2, x_1, x_2) \end{aligned}$$

and is nondecreasing in  $x_1$  and  $x_2$ , respectively, for  $x_1 > 0$  and  $x_2 > 0$ .

This can be proved using induction combined with value iteration [29]. We omit the details.

*Theorem 3:* The optimal DC scheduling policy is a monotonic threshold policy with respect to the queue lengths, i.e., if it is optimal to transmit to user  $i$  in state  $(r_1, r_2, x_1, x_2)$ , then it is optimal to transmit to user  $i$  in state  $(r_1, r_2, x'_1, x'_2)$  for  $x'_i > x_i$  and  $x'_j = x_j (j \neq i)$ .

This follows from Lemma 3. We omit the detailed proof.

Fig. 5 shows the optimal policy for two users, calculated via value iteration [29], and illustrates the monotonicity property in Theorem 3. Both users have the same cost functions  $U_1(x) = U_2(x) = x^{1.1}$ , and the initial queue lengths are  $A_1 = A_2 = 10$ . The channel parameters are  $(\eta_1, r_1^{\max}) = (0.04, 3)$  and  $(\eta_2, r_2^{\max}) = (0.1, 3)$ . Since user 1 has a better channel than user 2, in most cases, user 1 has higher priority than user 2.

Although here we only consider a two-user system, we have observed that the property stated in Theorem 3 applies to the  $M(> 2)$ -user systems simulated in our numerical studies.

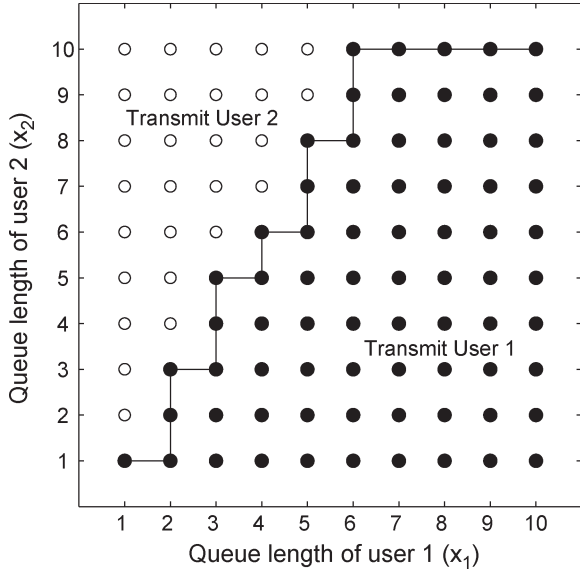


Fig. 5. Optimal scheduling policy as a function of the queue lengths for two users in the DC problem. A dot (circle) means it is optimal to transmit the HoL packet for user 1 (user 2).

### VI. NUMERICAL RESULTS

In this section, we compare the optimal LPA and DC scheduling policies with three simple policies, which select the HoL packet of user  $i^*$  as follows.

- 1)  **$U'R$  rule:**  $i_{U'R}^* = \arg \max_{1 \leq i \leq N} U'_i(x_i(n)) [1 - g_i(r_i(n))]$ , where  $U'(\cdot)$  is the derivative of the cost function.<sup>6</sup> This rule takes into account both the user's marginal cost and the expected transmission rate, which depends on  $g_i(\cdot)$  [26].
- 2) **Max  $U'$  rule:**  $i_{\text{Max}U'}^* = \arg \max_{1 \leq i \leq N} U'_i(x_i(n))$ . This rule takes into account only the user's marginal cost, and ignores channel conditions and number of transmissions attempts. This could model a situation where the scheduler has no channel information available.
- 3) **Max  $R$  rule:**  $i_{\text{Max}R}^* = \arg \max_{1 \leq i \leq N} (1 - g_i(r_i(n)))$ . This rule maximizes the expected transmission rate without regard to relative costs.

Fig. 6 shows total average cost for the preceding policies, applied to the LPA problem, as a function of user 2's cost rate  $c_{2,r_2} \triangleq c_2$  (for each  $r_2$ ). Here, the cost rate for user 1 is  $c_{1,r_1} = 1$  for each  $r_1$ . The channel parameters are  $(\eta_1, r_1^{\max}) = (0.01, 3)$  and  $(\eta_2, r_2^{\max}) = (0.4, 3)$ , so that user 1 has a better channel than user 2.

In Fig. 6, the  $U'R$  rule performs nearly the same as the optimal rule. When  $c_2$  is small (close to  $c_1$ ), scheduling decisions are determined primarily by the difference in channel conditions. In this region, the Max  $R$  rule is nearly optimal, and the Max  $U'$  rule performs significantly worse (up to 20% higher cost). When  $c_2$  is large, scheduling decisions are determined primarily by the difference in holding-cost rates. In this region,

<sup>6</sup>For the LPA problem, where  $U_i(\cdot)$  is given by (3), we set  $U'_i(\cdot) = c_{i,r_i^{\text{HoL}}}$ ; this represents the decrement of cost by successfully transmitting and decoding the HoL packet.

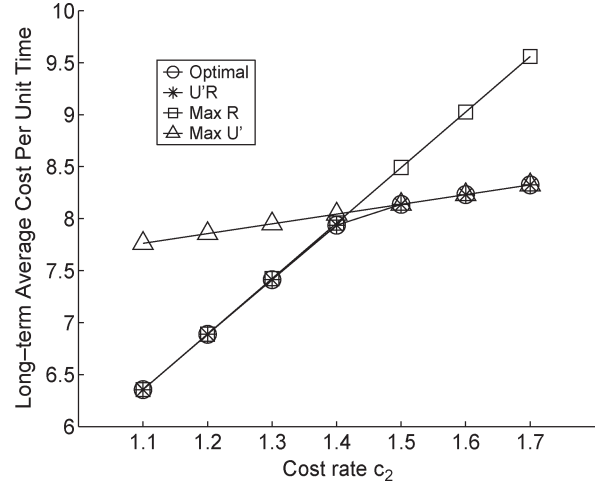


Fig. 6. Comparison of the optimal and heuristic scheduling policies in the LPA problem.

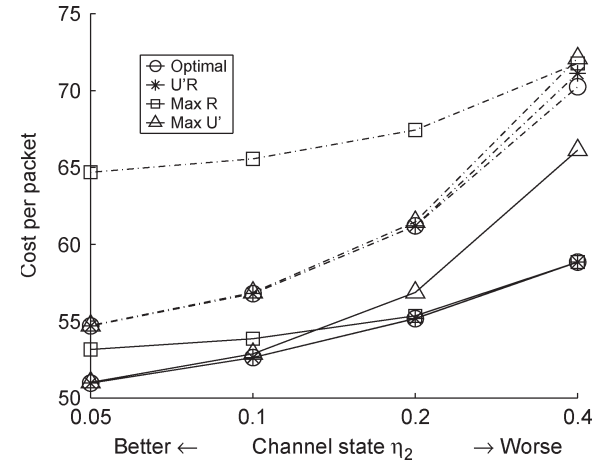


Fig. 7. Comparison of the optimal and heuristic scheduling policies in the DC problem (solid line:  $\kappa_2 = 1.08$ , dash-dotted line:  $\kappa_2 = 1.15$ ).

the Max  $U'$  rule is nearly optimal, while the Max  $R$  rule performs significantly worse (up to 15% higher cost).

To understand why the  $U'R$  rule performs well, consider standard ARQ, which is a special case of our problem with  $g_i(r_i) = g_i(0)$  for all  $r_i$  and  $r_i^{\max} = \infty$ . In this case

$$\begin{aligned} \frac{c_{i,r_i^{\text{HoL}}}}{T_{i,r_i^{\text{HoL}}}} &= \frac{c_{i,r_i^{\text{HoL}}}}{1 + \sum_{j=r_i^{\text{HoL}}}^{r_i^{\max}-1} \prod_{l=r_i^{\text{HoL}}}^j g_i(l)} = \frac{c_{i,r_i^{\text{HoL}}}}{\sum_{l=0}^{\infty} (g_i(0))^l} \\ &= c_{i,r_i^{\text{HoL}}} (1 - g_i(0)) = c_{i,r_i^{\text{HoL}}} (1 - g_i(r_i^{\text{HoL}})). \end{aligned} \tag{23}$$

Hence, according to Corollary 1, the optimal rule is exactly the  $U'R$  rule. For hybrid ARQ, this is no longer true in general, but Fig. 6 shows that the difference in performance is negligible.

Fig. 7 compares the optimal DC scheduling policy with the preceding heuristic policies. In this case, we plot the cost per packet versus channel parameter  $\eta_2$ . The cost functions are  $U_i(x_i) = x_i^{\kappa_i}$ , where  $\kappa_1 = 1.05$  and  $\kappa_2 \in \{1.08, 1.15\}$ . The channel parameters are  $(\eta_1, r_1^{\max}) = (0.01, 2)$  and  $r_2^{\max} = 4$ , i.e., user 1 has a better channel, but incurs less cost than user 2.



The initial queue lengths are  $A_1 = A_2 = 40$ . Results are shown for both values of  $\kappa_2$ .

Fig. 7 shows that the  $U'R$  rule performs quite close to the optimal policy. The relative performance of the other policies depend on the users' cost functions. When the cost functions are relatively close (e.g.,  $\kappa_1 = 1.05$  and  $\kappa_2 = 1.08$ ), scheduling decisions are determined primarily by the probability of decoding success. In this region, the Max  $R$  rule is nearly optimal (within 5%) and the Max  $U'$  rule performs significantly worse (up to 10% higher cost). On the other hand, when  $\kappa_1 = 1.05$  and  $\kappa_2 = 1.15$ , scheduling decisions are determined primarily by the difference between the cost functions. In this case, the Max  $U'$  rule is nearly optimal, and the Max  $R$  rule performs significantly worse (up to 18% higher cost).

## VII. CONCLUSION

We have considered channel-aware scheduling for wireless downlink data transmission with hybrid automatic repeat request (ARQ). An optimal scheduler minimizes the total average cost, where the cost function assigned to each user depends on the queue length and the number of transmission times for the HoL packet. We characterized the optimal scheduling policies in two situations by transforming these problems so that they fit into the Klimov framework. Namely, with linear cost functions and Poisson arrival processes, the optimal scheduling policy for the transformed problem is a fixed-priority policy. The priority indices can be computed in closed form, and increase with the number of unsuccessful transmissions. A different transformation is used for the draining problem with general increasing convex-cost functions. The optimal scheduling rule for the transformed problem is again a fixed-priority policy, but the priorities must be computed via Klimov's iterative algorithm. In that case, the priorities increase with queue length, and each packet is transmitted continuously until it leaves the system.

We also compared the optimal scheduler with a simpler myopic scheduling policy, the  $U'R$  rule, and showed that it is optimal without packet combining (standard ARQ). Through simulation, we found that the  $U'R$  rule performs very close to the optimal scheduler.

Our results assume that the scheduler knows the probability of a successful transmission. This is reasonable in slow-fading environments, where the channel is predictable over successive retransmissions, and in fast-fading environments where the channel statistics are stationary during and across transmissions. Further work is needed to extend these types of results to more general models of time-varying channels.

## APPENDIX PROOF OF COROLLARY 1

In LPAK, for queues  $(i, r_i)$  and  $(i, r'_i)$  with  $r_i < r'_i$ , by property 3) of Lemma 1 and (4),  $c_{i,r_i}/T_{i,r_i}^{(\Omega)} \leq c_{i,r'_i}/T_{i,r'_i}^{(\Omega)}$ . From Theorem 1, a packet in  $(i, r'_i)$  has priority over a packet in  $(i, r_i)$ , i.e., the priority of a packet is increasing with the number of transmission attempts. Thus, there can be at most one packet

with  $r_i > 0$  for each user  $i$ , and this packet has priority over all the user's other packets. This packet corresponds to the HoL packet in the LPA problem. Therefore, the optimal scheduling rule for LPAK is also optimal for LPA.

From Theorem 1, the queue with the highest ratio  $c_{i,r_i}/T_{i,r_i}^{(\Omega)}$  has the highest priority. By definition,  $T_{i,r_i^{\text{HoL}}} = T_{i,r_i^{\text{HoL}}}^{(\Omega)}$ , and so the HoL packet with the largest value of  $c_{i,r_i^{\text{HoL}}}/T_{i,r_i^{\text{HoL}}}$  among all the nonempty queues has the highest priority.

Let  $\Delta(r_i^{\text{HoL}}) = c_{i,r_i^{\text{HoL}}}/T_{i,r_i^{\text{HoL}}}$ . If  $r_i^{\text{HoL}}$  is replaced by  $r_i^{\text{HoL}'} > r_i^{\text{HoL}}$ , then  $\Delta(r_i^{\text{HoL}'}) \geq \Delta(r_i^{\text{HoL}})$ , whereas  $\Delta(r_j^{\text{HoL}'})$  stays the same for all  $j \neq i$ . Hence,  $\Delta(r_i^{\text{HoL}'}) \geq \Delta(r_j^{\text{HoL}'})$  for all  $j \neq i$ , i.e.,  $i$  has priority.

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