Abstract—LTE-Advanced networking standard introduces fixed relays as one of the ways to support the ambitious performance requirements for next-generation network services. The relays are introduced to increase the available capacity or to extend the coverage of the network. Balancing the dynamic interference scenarios and the complex resource management in relay-assisted networks presents one of the key challenges for the further development of LTE-A. This paper introduces the challenge of adaptive resource management and identifies simple adaptive scheduling methods based on channel quality information. The paper then utilizes a comprehensive simulation tool to show clear throughput benefits of using adaptive resource management for either the cell edge users or users experiencing very good signal-to-interference ratio, depending on the system objective.

I. INTRODUCTION

3GPP LTE-Advanced (LTE-A) introduces several new technical features which bring additional improvements in terms of coverage extension and/or increasing the capacity of the system. LTE-A enhances spectrum flexibility through carrier aggregation (CA) which gives the opportunity to use up to five components carriers. The release extends multi-antenna transmission where a maximum of 8 antenna ports is supported. For heterogeneous network deployments in terms of pico cells an additional feature to improve inter-cell interference coordination, as well as the support of relay nodes (RN) is introduced. Static relaying which can be used either for capacity or coverage extension is the focus of this paper. Several challenges have been identified to introduce the RNs to the LTE network [1]. They include maintaining backward compatibility and dealing with the higher dynamic of interference in the system. In addition it is important to design the scheduling of the available physical resources in the network which is the focus of this paper.

The RNs are classified according to their functionality. They have a wireless backhaul link (BL) which can be either deployed on co-channel (inband) or on dedicated channel (outband) [2]. An RN in its simplest mode has very little intelligence and just amplifies and forwards the received signal. This type of RN is defined as layer one RN. A layer two RN has at least the opportunity to decode and forward the signal to adapt modulation and coding schemes to the current channel quality. The radio resource control itself is controlled by the donor eNode-B (deNB) which serves the RN. Layer three RNs have full control of their radio resources and have the same functionality as an eNB. Within this study layer two RNs are used [3], [4]. The user equipment (UE) can be served via a direct link (deNB-UE) or via a two hop link which consists of the BL (deNB-RN) and the access link (RN-UE). In this context the advantage of a two-hop link is the split of direct links (DL) for UEs experiencing bad quality. A lower path loss (PL) on the access link (AL) is obtained since the distance to the serving RN is shorter than to the deNB. In addition the probability is increased to get line of sight conditions (LoS) for the AL. In case of a reuse factor of one the RNs as well as the eNBs have full access to the total system bandwidth.

However, to make use of that advantage in a scenario where inband BL is deployed it implies the need of excellent link quality to overcome the problem that there is usually a loss in time since the RN cannot receive and send simultaneously.

The 3GPP has defined the multicast broadcast single frequency network (MBSFN) subframe which is used for backhaul reception at the RN in [2]. As depicted in Fig.1 this special type of subframe gives the opportunity to transmit at first the physical downlink control channel (PDCCH) which is expected by each UE within each subframe and then switches to the reception mode where the RN can receive data from the deNB. During that time the UEs assigned to the RNs are not served with data. In the end of an MBSFN subframe the RN again switches back to sending mode. As mentioned before, the used approach to serve the inband RNs is half-duplex, which brings a loss in time. Moreover there exist some additional constraints which are summarized as follows.

In the LTE frequency division duplex (FDD) downlink frame structure several subframes are not configurable as MBSFN subframes since they transport system relevant information such as physical broadcast channel (PBCH) (1st subframe) or primary and secondary synchronization signal

![Figure 1. Use of MBSFN subframe structure to receive data at the RN](image-url)
A uniformly UE distribution is considered. The assignment of RN and the receiving antenna of the UEs are omnidirectional.

Horizontal beam width of 40

In Fig. 2 the used RN extended LTE-A network with 19 hexagonal cell sites and 3 sectors per site is shown. In each sector 4 RN are placed near the cell edge with planning gain based on [9]. Each cell sector is equipped with a directional antenna with a 3dB horizontal beam width of 70°. The receiving antenna of the RN has a directional antenna with a 3dB horizontal beam width of 40°. Transmitting antennas of the RN and the receiving antenna of the UEs are omnidirectional. A uniformly UE distribution is considered. The assignment of UEs is done based on the reference signal received power (RSRP). Then an estimation is done considering the worst case wSINR for all UEs where all RNs are transmitting. The possible DLs are compared to the AL of the two-hop link. Out of the wSINR a supportable rate (SR) is derived based on a Shannon bound approximated curve [10]. If all SR are known initially, the BL and the AL of the two hop link, are compared with each other. The minimum of both values is than compared with the SR of the DL. Therefore the UEs will only be served on the physical resource blocks (PRB) of the RN if the quality is better than on the DL.

B. Dynamic Interference, Scheduling and Buffer

The available system bandwidth is separated in PRBs, which can be assigned to UEs every ms by the scheduler. The RNs have the same available system bandwidth as the deNB. The interference situation can change rapidly by switching RNs from receiving into sending mode in different subframes than in the neighboring sites or sectors. Even if the same subframes are configured as MBSFN, the amount of transmitted PRBs could be different and thus interference will be more dynamic in frequency and time than in homogeneous networks. The scheduling can be shifted from a fair resource distribution, which would be a RR approach without considering the channel quality, to a max-min scheduler where the UEs with the best link quality get all the resources. A PF scheduler prioritizes the UEs with good quality conditions while also serving UEs with worse channel quality.

Within the study a modified RR as well as modified PF scheduling approaches are considered. Therein the well known metrics are extended by a weighting factor which prioritizes RNs to get data when they are able to receive during MBSFN subframes. Hence, it is not a full buffer, fully loaded system for the AL. Limited by an introduced buffer scheme at the RN the data is scheduled based on the common schemes at the RN. The maximum configurable number of 6 MBSFN subframes within a radio frame is used. The RNs are only transmitting during non-MBSFN subframes. For each UE the SINR on its scheduled PRBs is calculated. MBSFN subframes. Hence, it is not a full buffer, fully loaded system for the AL. Limited by an introduced buffer scheme at the RN the data is scheduled based on the common schemes at the RN. The maximum configurable number of 6 MBSFN subframes within a radio frame is used. The RNs are only transmitting during non-MBSFN subframes. For each UE the SINR on its scheduled PRBs is calculated. MBSFN subframes are configured synchronous to prevent even higher dynamics in interference. This is done to have less influencing variables on the results, therefore having a better impression of the scheduler’s behavior and on the influence of the different weighting factors.

An additional challenging problem is to handle the channel estimation feedback in such dynamic systems [11] which will be investigated in the future (see Section V).

III. ADAPTED BASIC SCHEDULING SCHEMES BY DIFFERENT TYPES OF WEIGHTING FACTORS

A. Weighted Round Robin Scheduler

The common round robin approach is defined by,

$$P_{PRB} = \left[ \frac{p_{RR}(u)}{\sum_{i=1}^{U} p_{RR}(i)} \right] \cdot \text{PRB}_{\text{total}} \right), \quad (1)$$

where $u$ is the user index, $P_{PRB}$ is the probability of a user being scheduled, $p_{RR}(u)$ is the probability of user $u$ being scheduled, and $\text{PRB}_{\text{total}}$ is the total number of PRBs available.

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where the number $m$ of PRB which will be assigned to the user $u$ is defined as the priority $PRB$ of the user which is divided by the sum of the total number of priorities of the assigned users, multiplied with the maximum number of available $PRB_{total}$. The priority factor $PRB$ can be defined as one which results in a common round robin approach. This is done by,

$$
PRB(u) = 1, \quad \forall u \in U;
$$

where the same priority $PRB$ for user $u$ of all users $U$ is given. To prioritize the RN at the dENB,

$$
P_{PRB}(u) = 1 \cdot w_{PRB}(u),
$$

$$
 w_{PRB}(u) = \left\{ \begin{array}{ll} \frac{1}{w(rn)} & \text{for UE} \in \text{RN} \text{ with } w(rn) = U_{served}, \\
\end{array} \right.
$$

is used. Here the weighting factor $w$ is given. If the user $u$ is defined as an RN the number of UEs assigned to the RN ($U_{served}$) equals the weighting factor $w(rn)$. In all other cases $w_{PRB}$ is defined as one.

**B. Weighted Proportional Fair Scheduler 1**

As a basis for the second alternative in this study the common PF scheduling approach is given by,

$$
PF1(u, PRB) = w_{PF1}(u) \cdot \frac{SR(u)}{\sum_{i} THRU_{i}},
$$

The priority is calculated for each user $u$ on each available PRB based on Eq. 4 where the $SR$ of each user $u$ is calculated as explained in Section II. The exponent $\alpha$ is defined as the fairness factor and set to one to keep proportional fairness.

As an example, if $\alpha$ would be set to zero a max-min scheduling approach would be used which can be seen as unfair since all the resources would be always assigned to users with the best $SR$.

The weighting factor $w_{PF1}$ is calculated in the same manner as for the modified approach in Eq. 3. The throughput $THR$ of each user $u$ at every time $t$ is calculated by,

$$
THR(u, t) = \beta \cdot THR(u, t - 1) + (1 - \beta) \cdot d(u, t),
$$

The first term is defined as the previous user throughput already transmitted. The second term $d$ is defined as the instantaneous throughput transmitted in the latest time instance. The factor $\beta$ is defined as the forgetting factor which influences the priority based on a weighting of past transmitted data.

As an example, if $\beta$ would be defined as one only the initial transmitted throughput would influence the metric, thus it would result in a max-min scheduling approach. During this study the forgetting factor $\beta$ is set to 0.97.

This approach does not take into account the link quality which will be introduced due to the following.

**C. Weighted Proportional Fair Scheduler 2**

As a second alternative for the weighted PF scheduling a different weighting factor is defined by,

$$
PF2(rn) = \left\{ \begin{array}{ll} w_{PF1}(rn) \cdot w_{PF2}(rn) & \text{if } w_{PF2}(rn) > 1, \\
 w_{PF1}(rn) & \text{if } w_{PF2}(rn) \leq 1,
\end{array} \right.
$$

and

$$
w_{PF2}(rn) = \sum_{i} \left( \frac{SR_{AL}(i)}{SR_{BH}(rn)} \cdot \frac{SR_{AL}(i)}{\sum_{j} U_{served} \cdot SR_{AL}(j)} \right) = \sum_{i} \left( \frac{SR_{AL}(i)^{2}}{SR_{BH}(rn) \cdot \sum_{j} U_{served} \cdot SR_{AL}(j)} \right),
$$

Where $w_{PF}$ is the weighting factor for the RNs, $w_{PF1}$ is the factor given by the number of users as in Eq. 4 and $w_{PF2}$ is the factor introduced by comparing the $SR$ of access and backhaul links given in Eq. 7.

A ratio between access (SR$_{AL}$) and backhaul (SR$_{BH}$) links is calculated. If the AL has a better quality than the BH link, $w_{PF2}$ is larger than one and thus more resources should be spent to serve the RN than in the previous approach in Section III-B. The ratio between AL and BH link is multiplied by the ratio between the quality of the AL and the sum of all ALs of the RN which reflects the importance of the link among all connected users ($U_{served}$) to the RN. The resulting product of multiplying the two ratios in sum over all served users is defined as $w_{PF2}$.

**D. Weighted Proportional Fair Scheduler 3**

As described in Section I it is of greatest importance to provide excellent link conditions for the BH link because of the loss in time by using inband RNs. The proposed weighting factor in Section III-C might remain as the number of users, since the BH is not worse in many cases compared to the AL.

For that reason, another alternative is defined. This approach relates the AL and BH links by comparing the theoretical amount of data that each link can provide given an initial resource distribution. If the RN receives less data than the users demand, the weighting factor increases the scheduling priority of the RN. If the UEs do not require as much data as the BL can offer the weighting factor is less than unity to allow other RNs or UEs at eNBs to receive more data. Based on the $SR$ a proportional distribution is done in order to determine the amount of PRBs per UE. As given by,

$$
d(u) = PRB_{AS}(u) \cdot SR(u) \cdot BW_{PRB} \cdot T_{SF}.\n$$

The assigned resources per UE, $PRB_{AS}(u)$ are multiplied with the $SR$, the $BW$ of a PRB and the duration of a subframe $T_{SF}$ to calculate the amount of data which has to be transmitted for a single UE. Finally the ratio between access $d_{AL}$ and backhaul $d_{BH}$ data, respectively is calculated to obtain the weighting factor for RNs as described by,

$$
w_{PF3}(rn) = \sum_{i} \frac{d_{AL}(i)}{d_{BH}(rn)}.
$$

**IV. SYSTEM LEVEL SIMULATION ASSUMPTIONS AND RESULTS**

In Tab. I the most important simulation settings are summarized which follow [9].

In Fig. 3-5 the simulation results are depicted for all approaches as cumulative distribution functions (CDF). On the
left side the wSINR and on the right side the UE throughput are shown. Fig. 3 gives results for the UEs which are served by the macro via the DL. Fig. 4 illustrates the results for the UEs which have been served by the RNs and finally the overall performance is depicted in Fig. 5. In addition it is worth to mention that in all scenarios approx. 50 per cent of all UEs are served by RNs and the other half through DL connection. The black curve in all CDFs is always defined as the reference scenario where no RNs are deployed in the system. 

### B. UEs served through RNs (UE@RN)

In Fig. 4 a deviation in wSINR compared to the reference scenario can be observed. The interference experienced by this user group is higher than in a macro only case, since all the RNs are transmitting simultaneously (assuming they have data to forward). Therefore, the wSINR can be even worse than in a macro only case. However, in situations where the interference is not significant, higher SINR values than for macro UEs are possible as a result of the shorter separation distances between UEs and the RNs. 

If no weighting factor is applied in order to prioritize the data transmission to the RNs, UEs connected via a two hop link are not scheduled often enough due to the time division scheme applied to manage the RNs. Thus, the UE throughput of the basic RRS is worse than the reference scenario. Otherwise, when applying weighting factors, the throughput is significantly improved, especially for UEs with very good wSINRs (over the 90th percentile). That is the case of the PFS schemes. A higher impact can be seen of the schedulers at low throughputs (5th percentile) in comparison to the DL users. As mentioned for the wSINR, RN users suffer higher interference and the majority of them is expected to be mainly localized in that part of the throughput CDF. 

### C. Overall served UEs (UE@MBS+RN)

Fig. 5 summarizes the results for all UEs (direct and two hop links). Three regions can be observed at the throughput CDF: from 0 to 60th percentile, there is a higher influence of the RN users; therefore, schedulers which provide more data to the RNs (PFS-1 and PFS-2) have the best performance. In the case of PFS-3, the amount of remaining resources to be used by the macro UEs is too low, counteracting the benefit of the weighting factor in the overall system. From 60th to 95th percentile, the higher influence of the macro UEs can be seen by taking into account the best result provided by the basic RR scheduler, which benefits the most these users. Finally, from 95th to 100th percentile, UEs with the best wSINR are represented. As mentioned before, those UEs are served through the RNs and are better served when providing enough data, which is possible by using the PFS-3 alternative. Finally it can be seen that the used metrics can either improve the capacity (avg. gain) or the coverage (5 percentile). 

### V. Conclusion and Future Work

This paper shows simulation results for a RN enhanced LTE network. It defines and analyzes different approaches of how RNs can be scheduled at the MBS, taking into account the usage of MBSFN subframes. It can be seen that the definition of how to prioritize the RNs in the scheduling process is important. The goal could be a capacity extension or a coverage improvement for the system. Five possible metrics are compared and the results show either a coverage improvement or a capacity extension. It could also be observed of cases where the ALs are better than the backhaul, as a result of the site planning.
that the wideband SINR conditions of the users are influenced by the different scheduling approaches.

The next step will be to introduce a MIMO system with a frequency selective channel model and channel feedback information, considering [11]. Based on that a deeper analysis can be done of how to handle a fast outdated of the channel estimation feedback information in such a system with high dynamic of interference which needs to be investigated further, as recommended in [1], [12].

REFERENCES