LTE Part 2: Data Link and Physical Layers

Physical Layer based on:

David Astély, Erik Dahlman, Anders Furuskär, Ylva Jading, Magnus Lindström, and Stefan Parkvall, Ericsson Research, *LTE: The Evolution of Mobile Broadband*, IEEE Communications Magazine, April 2009. pp.44-51

Scheduling from:

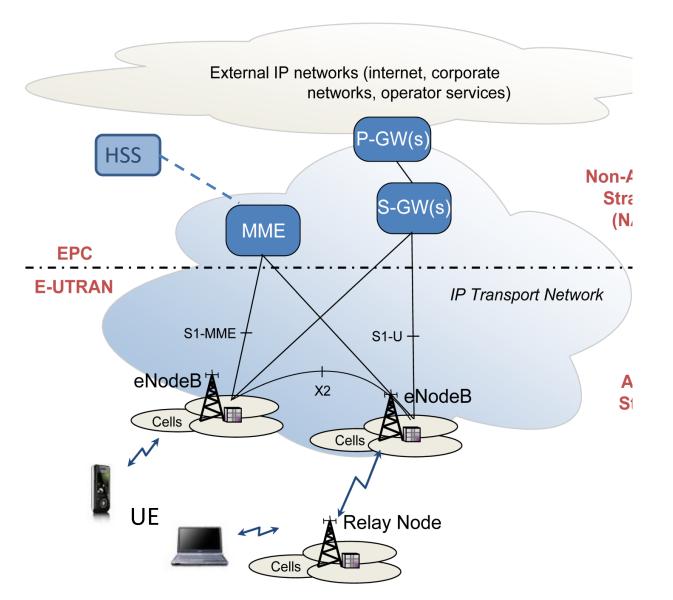
Anna Larmo, Magnus Lindström, Michael Meyer, Ghyslain Pelletier, Johan Torsner, and Henning Wiemann, Ericsson Research: *The LTE Link-Layer Design*, IEEE Communications Magazine, April 2009. pp. 52-59

Additional reading:

C Beard & W. Stalling, Wireless Networks and Systems, 2nd edition, Chapter 14, pp. 470-509

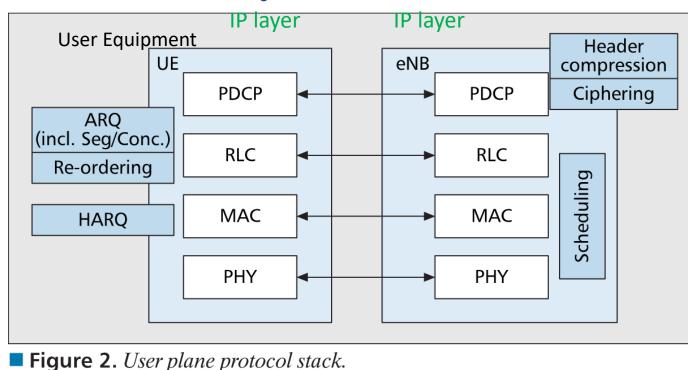
LTE: EPC + E-UTRAN architecture

Evolved Packet Core + Evolved Universal Terrestrial Radio Access Network



- Flat all-IP multiaccess core network
- Enhanced NodeB
 (eNB) (former base
 station/NodeB)
- **UE** User Equipment
- MME Mobility
 Management Entity
 (Control Plane)
- S-GW Serving Gateway (User Plane)
- P-GW Packet Network Gateway
- HSS Home
 Subscriber Server

Link Layer – User Plane Protocol Stack



The LTE link layer consists of three sublayers:

- The Packet Data Convergence Protocol (**PDCP**)
- The radio link control (**RLC**)
- The medium access control (MAC)

PDCP is responsible for IP header compression and ciphering.

- Supports lossless mobility in case of inter-eNB handovers
- Provides data integrity protection to higher layer protocols.

RLC comprises: ARQ (Automatic Repeat reQuest) functionality and

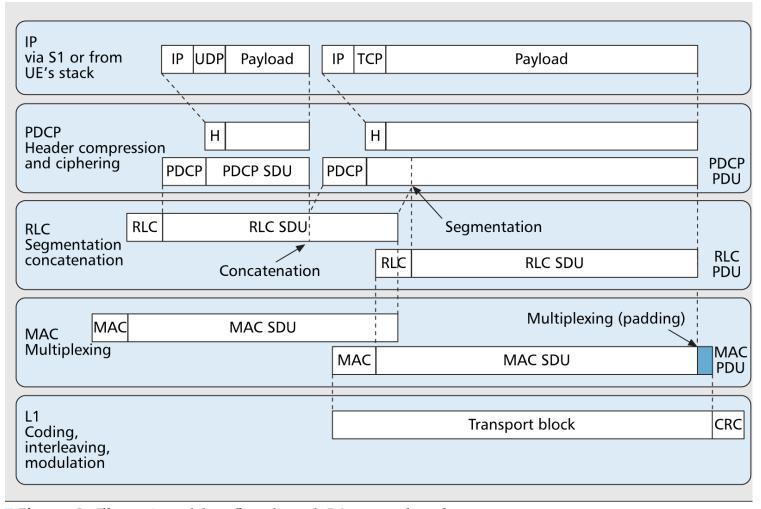
supports data segmentation and concatenation

MAC provides HARQ (Hybrid ARQ) and is responsible for

- medium access control
- scheduling operation and random access.

Peer sublayers in UE and eNB are logically connected.

Data Flow Through L2 Protocol Stack



■ Figure 3. *Illustration of data flow through L2 protocol stack.*

The data flow of an IP packet through the link-layer protocols down to the physical layer (L1). The figure shows that each protocol sublayer adds its own protocol header to the data units.

Protocol Data Units (PDUs)

- Each sublayer of the link layer operates with its own PDU consisting of a header and Service Data Unit (SDU)
- The Packet Data Convergence Protocol (PDPC) header has a D/C bit distinguishing Data from Control frame and the Sequence Number (SN) field of the varying length (5, 7, 12 bits)
- The Radio Link Control (RLC) PDU has four basic formats
 - Transparent Mode Data (TMD) PDU
 - Unacknowledged Mode Data (UMD) PDU
 - Acknowledged Mode Data (AMD) PDU
 - Status PDU
- The Medium Access Control (MAC) PDU can multiplex data from different channels, therefore, the header contains sub-headers control elements for a collection of multiplexed SDUs
- The Physical Layer (L1) operates with a Transport Block related to the time-frequency resource block

Error Handling 1

- As in any communication system, there are occasional data transmission errors, due to noise, interference, fading, etc.
- In mobile/wireless networks one in ten data blocs can be corrupted.
- Link-layer, network-layer (IP), and transport-layer protocols are not prepared to cope with bit errors in headers, and the majority of the protocols are not also capable of handling errors in the payload.
- Therefore, a fundamental design choice for LTE has been
 - not to propagate any bit errors to higher layers but rather
 - to drop or retransmit the entire data unit containing bit errors.

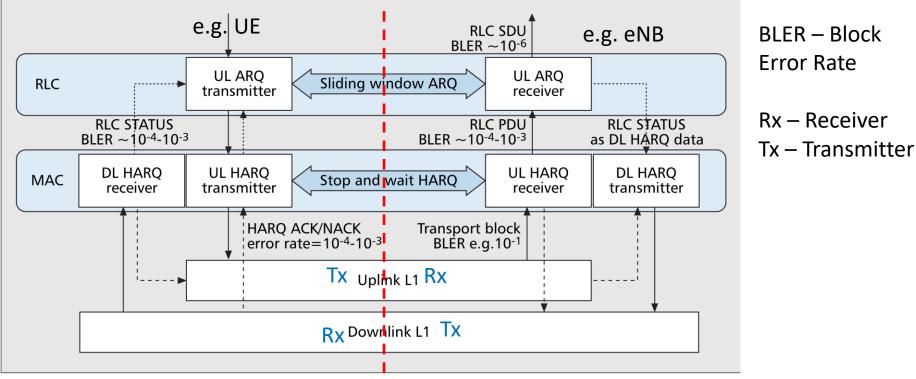
Error Handling 2

- The physical layer uses Forward Error Coding (Turbo codes) to correct as many errors as possible.
- The 24-bit CRC attached by the physical layer allows detecting bit errors by the MAC layer
- MAC layer uses the mechanisms of fast HARQ (Hybrid Automatic Repeat reQuest)
- The RLC layer uses the mechanism of the Sliding Window ARQ
- As the result, very few errors slips through the above mechanisms and reaches the IP layer

Hybrid ARQ – HARQ

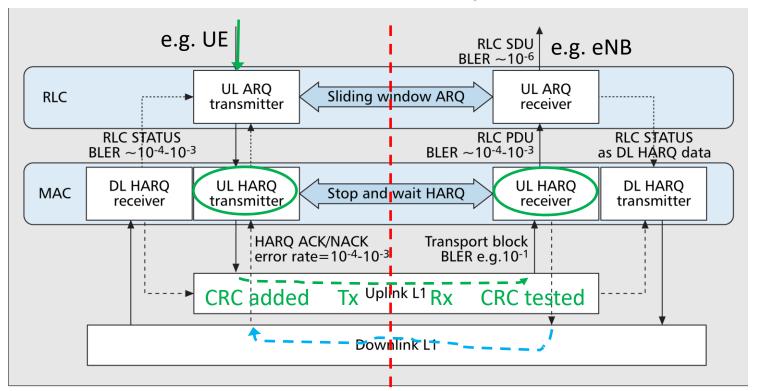
- The HARQ scheme on the MAC sublayer performs retransmissions of corrupted transport blocks and thereby corrects the majority of all transmission errors (from 10^{-1} down to 10^{-3})
- The HARQ mechanism uses **multiple stop-and-wait** HARQ processes.
- HARQ uses a single-bit feedback ACK/NACK
- Short **round-trip time** of only 8ms to provide information about the successful transmission.
- Multiple stop-and-wait processes allow continuous transmission, which cannot be achieved with a single stop-and-wait scheme.
- The resulting functionality and performance is comparable to that of a **window-based selective repeat** protocol.

HARQ and ARQ retransmissions on MAC and RLC layers



- Figure 4 HARQ and ARQ retransmissions on MAC and RLC layer.
- The figure shows the RLC, MAC and L1 layers in the User Equipment (UE) and the Evolved Node B (eNB) aka the base station
- The **physical transmission** occurs in the L1 layer:
 - Uplink (UL) from UE to eNB, and Downlink (DL) from eNB to UE
 - From a related Transmitter (Tx) to the Receiver (Rx)
- The RLC and MAC layers communicate logically with the equivalent peer layers through the logical Tx and Rx

HARQ ACK/NACK



BLER – Block Error Rate

- Figure 4 HARQ and ARQ retransmissions on MAC and RLC layer.
- Assume that data is **uploaded** from the UE to the eNB:

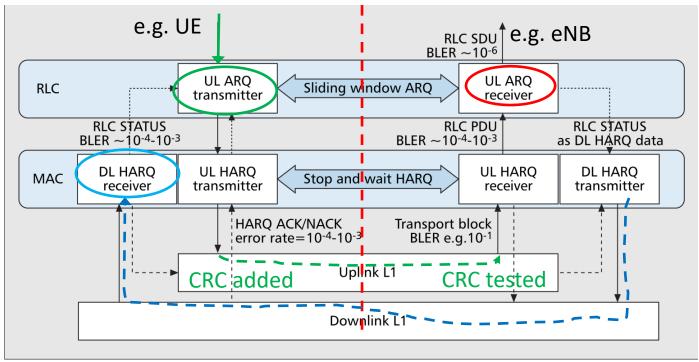
RLC: UL ARQ $Tx \longrightarrow MAC$: UL HARQ $Tx \longrightarrow L1$: UL Tx

- The CRC is added by the physical layer (L1) of the UL Tx and tested by the L1 of the Rx. (green line) $BLER = 10^{-1}$
- The Rx MAC layer (UL HARQ Rx) sends ACK or NACK through the "DL L1" back to Tx MAC layer (UL HARQ Tx) (blue line) ACK/NACK error rate $\approx 10^{-3}$
- In the case of NACK, the MAC PDU is re-transmitted (green again)

From HARQ to ARQ

- In the single-bit HARQ feedback the probability for misinterpreting a negative acknowledgment (NACK) as a positive acknowledgment (ACK), and thereby causing a residual packet loss, is in the order of 10^{-4} to 10^{-3} .
- Such residual loss rates is too high to achieve high-data rates at the TCP level.
- Certain errors in other control signalling, such as scheduling information, also result in HARQ failures.
- Therefore, the fast HARQ protocol with low-overhead, ACK/NACK feedback and retransmissions with incremental redundancy
- is complemented by a highly reliable window-based selective repeat-ARQ protocol that resides in the RLC sublayer.
- PDUs are numbered and re-transmitted if error is detected within a specified sliding window.

ARQ ACK n / NACK m



BLER – Block Error Rate

- Figure 4 HARQ and ARQ retransmissions on MAC and RLC layer.
- Assume that data is uploaded from the UE to a eNB and the error in RLC PDU is detected by the UL ARQ receiver. BLER $\approx 10^{-3}$
- In the sliding window ARQ the data is acknowledged up to PDU # n inside the agreed sliding window.
- If there is an error in the RLC **PDU** #m, this PDU needs to be re-transmitted.
- The RLC transmission status is sent as data from DL HARQ Tx to DL HARQ Rx and if it is correctly received, the status is sent to UL ARQ Tx which re-transmits the PDU #m

HARQ and ARQ retransmission (1)

- All **re-transmissions are timed** to make sure that operations are completed in the set amount of time.
- If the RLC receiver detects a gap in the sequence of received PDUs based on the RLC sequence number, it starts a reordering timer assuming that the missing packet still is being retransmitted in the HARQ protocol.
- The reordering functionality reuses the same RLC sequence numbers as the ARQ mechanism
- In the rare case that the reordering timer expires:
 - an RLC acknowledged-mode (AM) receiver sends
 a status message to its transmitting peer entity
 - comprising the sequence number of the missing RLC
 PDU(s)

HARQ and ARQ retransmission (2)

- The MAC layer treats the RLC status message as any other data, meaning that it also applies the same HARQ operation and CRC to this message.
- Consequently, errors or loss of the ARQ feedback can be detected and recovered by sending another RLC status.
- HARQ retransmissions correct most errors (from 10^{-1} to 10^{-3}).
- ARQ retransmissions BLER reduces further down to 10^{-6}
- Together, the BLER has been reduce from 10^{-1} in the original transport block to only 10^{-6} in the resulting RLC SDU.
- Such amount is acceptable to the TCP layer.

Unacknowledged Mode (UM)

- Services that can sustain error rates on the order of 10^{-3} to 10^{-2} , while gaining from reduced delays,
- can be mapped to a radio bearer running the RLC protocol in **unacknowledged mode** (UM), that is, without the second ARQ layer.
- In that case, residual errors on the MAC layer are not recovered, but packet losses propagate to higher layers.
- RLC UM generally is assumed to be used for VoIP and real-time gaming traffic.

Handover support (1)

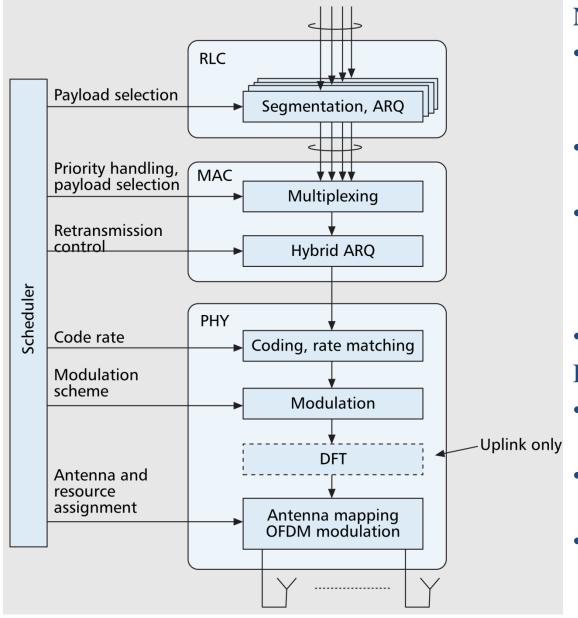
- In LTE, the UE performs measurements when radio conditions reach a certain configured threshold and provides measurement reports to the eNB it is connected to.
- The involved eNBs must negotiate through the X2 interface and decide whether or not to handover a UE to another cell or eNB.
- The EPC (evolved packet core) is not involved in the preparation signalling unless the change of serving cell involves an S1 handover as well.
- In the case of inter-eNB handover, the source eNB prepares neighboring eNBs over the X2 interface and then transmits a handover command to the UE with the required information to perform the handover to one of the prepared eNBs.
- The source eNB can forward data to the target eNB.

Handover support (2)

- For RLC AM (Acknowledge Mode) data bearers, the source eNB forwards unacknowledged downlink PDCP PDUs with their sequence number (SN) and not-yet-transmitted IP packets received over the S1 interface to the target eNB.
- It also can forward the uplink-PDCP-service-data units (SDUs) received out-of-sequence to the target eNB.
- For RLC UM data bearers in the downlink, only the not-yettransmitted IP packets received from the S1 interface are forwarded.
- The PDCP layer ensures that no data is lost at handover for RLC AM bearers by retransmitting missing data.
- PDCP in the UE also handles duplicate removal and insequence delivery of the PDCP SDUs received from the source eNB and from the target eNB based on the PDCP SN (Sequence Number)
- For RLC UM, no data is retransmitted by the PDCP.

LTE Layers Controlled by the Scheduler

From PDCP (Packet Data Convergence Protocol



Note:

- the basic functions of the scheduler located at the eNB (base station)
- some details of the Physical Layer:
- Bit coding includes: QPSK, 16QAM and 64QAM, followed by OFDM modulation
- Subcarrier spacing is 15kHz Bit processing includes:
- A 24-bit cyclic redundancy check (CRC),
- Forward Error Coding in the form of <u>turbo-coding</u>,
- Scrambling, puncturing, interleaving similar to 802.11

Physical Layer Characteristics (1)

The physical layer transport block, is protected by

- A cyclic redundancy check (24-bit CRC),
- Forward Error Coding in the form of <u>turbo-coding</u>,
- Scrambling, puncturing, interleaving is similar to 802.11

The basic LTE radio **resource block** is an addressable **time-frequency grid** that:

- Assembles 12 15kHz subcarriers and has a bandwidth of 180 kHz.
- Has a subframe duration of only 1 ms.
- Such a short subframe enables the exploitation of channel variations by **scheduling users** depending on their **current channel quality**.
- It allows a short **H**ybrid **A**utomatic **R**epeat re**Q**uest (**HARQ**) round-trip time of only 8ms.

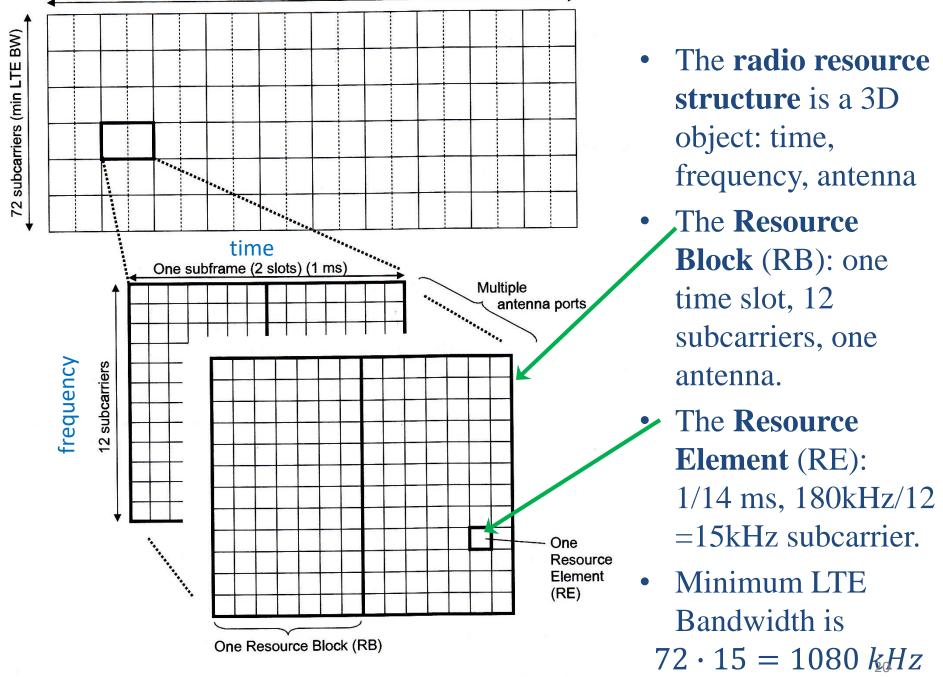
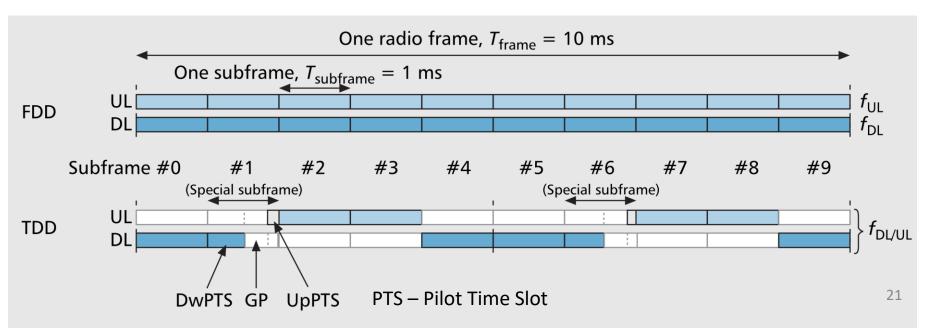


Figure 6.1: Basic time-frequency resource structure of LTE (normal cyclic prefix case).

One radio frame = 10 ms

Frames

- The transmitted signal is organized into **subframes** of 1ms duration, each consisting of 14 or 12 OFDM symbols, depending on whether normal (4.7μs) or extended (16.7μs) **cyclic prefix** (time guard) is used.
- Ten subframes form a 10ms radio frame.
- The short subframe duration of 1ms results in small delays for:
 - user data,
 - control signalling such as the HARQ feedback



One radio frame, $T_{\text{frame}} = 10 \text{ ms}$ One subframe, $T_{\text{subframe}} = 1 \text{ ms}$ UL **FDD** DL and Subframe #0 #2 #3 #4 #7 #8 #5 #6 #9 (Special subframe) (Special subframe) UL $f_{\mathsf{DL}/\mathsf{UL}}$ DL PTS – Pilot Time Slot, GP – Guard Period DwPTS GP UpPTS

- LTE supports two duplexing schemes:
 - FDD (frequency division duplex)
 - TDD (time division duplex) aka TD-LTE.
- The frame structure is almost identical for both duplexing schemes and can be handled by one chip set.
- In TD-LTE a special subframe with pilots is added to provide the required guard time for downlink-to-uplink switching
- FDD/TDD deployment issues are discussed here

LTE Physical Transmission Formats

- Channel coding
 First add CRC codes of 8, 16, or 24 parity bits.
- Modulation
 Supports downlink and uplink QPSK, 16QAM, and 64QAM, depending on channel conditions and UE capabilities. Use 4-Bit CQI (see Table 14.7 of textbook, p.497)
- Scrambling
- Reference Signal
 Use for measuring channel conditions
- H-ARQ

Turbo code first applied to code block. Any transmission will use H-ARQ at receiver to combined new data with previously received block.

Table 14.7 4-Bit CQI Table

CQI Index 60	Modulation	Code Rate x 1024	Efficiency (bit/Hz)
0		Out of range	
1	QPSK	78	0.1523
2	QPSK	120	0.2344
3	QPSK	193	0.3770
4	QPSK	308	0.6016
5	QPSK	449	0.8770
6	QPSK	602	1.1758
7	16QAM	378	1.4766
8	16QAM	490	1.9141
9	16QAM	616	2.4063
10	64QAM	466	2.7305
11	64QAM	567	3.3223
12	64QAM	666	3.9023
13	64QAM	772	4.5234
14	64QAM	873	5.1152
15	64QAM	948	5.5547

LTE radio resource block showing scheduling

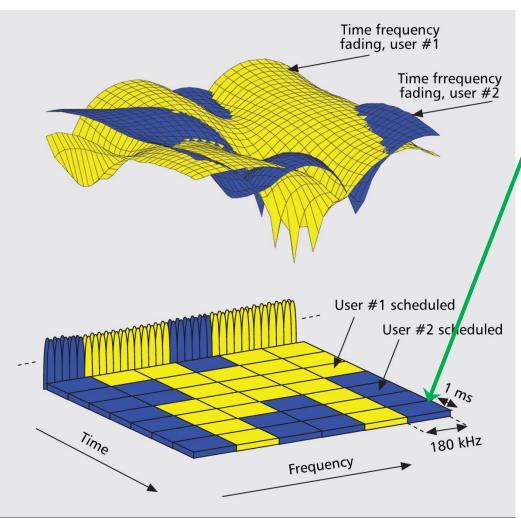


Figure 3. *Channel-quality variations in frequency and time.*

- The **radio channel quality** varies with time and frequency.
- In the example the scheduler allocate the radio resource
 blocks
- The basic **radio allocation unit** is a 1ms subframe with 12
 15kHz subcarriers = 180kHz
- The scheduler can re-allocate the radio resources every radio frame (10ms)
- Some blocks carry the control information.

Allocating the resource blocks

• A short subframe enables the exploitation of channel variations by **scheduling users** depending on their **current channel quality**.

Wide range of user-data rates is available by:

- Allocating a variable number of resource blocks to a user
- Selecting a modulation and coding scheme to meet the current channel conditions
- Selecting scalable transport block sizes.
- ➤ Data rates can be increased further by aggregating up to **eight radio streams** by utilizing multiple-input multiple-output (MIMO) transmissions

Scheduling Request Mechanism

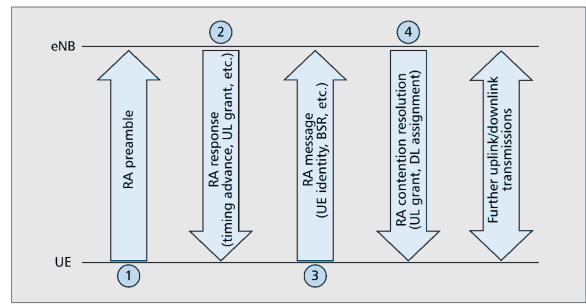
- To allow the UE to request uplink-transmission resources from the eNB, LTE provides a **Scheduling Request** (SR) mechanism.
- The SR conveys a single bit of information, indicating that the UE has new data to transmit.
- Two types of the SR mechanism:
 - dedicated SR (D-SR), where the SR is conveyed on a dedicated resource on the Physical Upload Control
 Channel (PUCCH)
 - random access-based SR (RA-SR), where the SR is indicated by performing an RA (Random Access) procedure.
- The D-SR is simpler than the RA-SR but assumes that the uplink of the UE already is time aligned.

Random Access (1)

- To keep transmissions from different UEs orthogonal, uplink transmissions in LTE are aligned with the frame timing at the eNB.
- A random access (RA) procedure is performed to acquire time alignment when:
 - timing is not aligned yet,
 - or alignment was lost due to a period of inactivity during which time alignment was not maintained by the eNB,
- The RA procedure establishes uplink time alignment by means of a four-phase contention based procedure :

Random Access Procedure

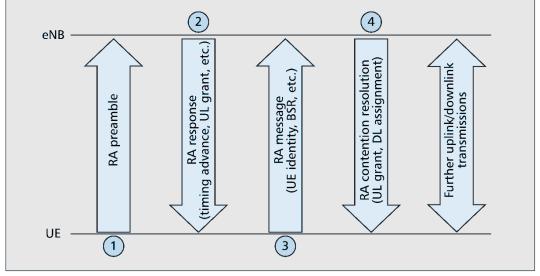
RA Preamble (1)



■ Figure 5. Contention-based random access procedure.

• The UE randomly selects an RA preamble sequence from the set of sequences available in the cell and transmits it on an RA channel.

RA Response (2)

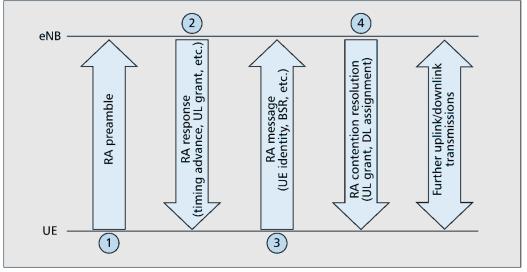


■ Figure 5. Contention-based random access procedure.

The eNB detects the preamble transmission,

- estimates the uplink transmission timing of the UE,
- responds with an RA response providing the UE with
- the correct timing-advance value to be used for subsequent transmissions and with a first grant for an uplink transmission.

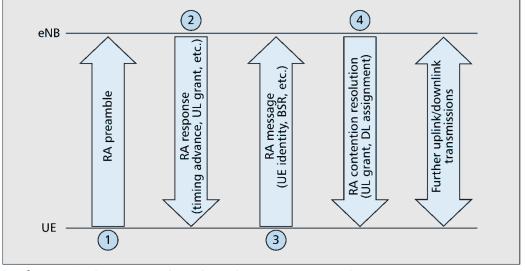
RA Message (3)



■ Figure 5. Contention-based random access procedure.

- The randomly selected RA preamble does not enable unique identification of the UE,
- it is possible that multiple UEs attempted RA with the same RA preamble sequence on the same RA channel
- The UE provides its identity to the eNB with the first scheduled uplink transmission.

RA Contention Resolution (4)



■ **Figure 5.** *Contention-based random access procedure.*

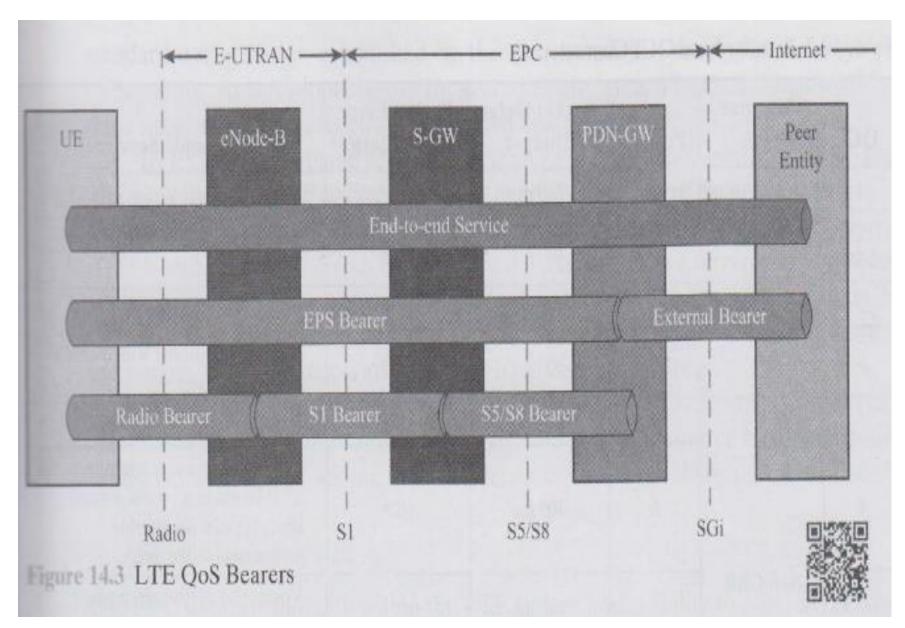
- The eNB receives the RA message transmitted in phase 3
- Only one RA message is typically received even if two or more were transmitted by contending UEs.
- The eNB resolves the (potential) contention by echoing the received UE identity back.
- The UE, seeing its own identity echoed back, concludes that the RA was successful and proceeds with time-aligned operation.

More on the Random Access Procedure

- UEs that do not receive an RA response or do not receive their own identity in the contention resolution, must repeat the RA procedure.
- In the case of congestion, the eNB can provide a back-off indicator to instruct UEs that did not succeed with their RA attempt to apply a back-off procedure.
- The back-off indicator is multiplexed with the RA responses.
- Note that for cases where an RA is anticipated by the network, that is, at handover completion and eNB-triggered uplink re-alignment, LTE also provides a faster two-phase contention free RA procedure.
- In this case the eNB assigns a dedicated preamble to be used by the UE.
- Because the UE that corresponds to the received dedicated preamble is known, phase 3 and 4 are not required.

More on Scheduling Request

- If the uplink of the UE is not time aligned, RA-SR must be used to (re-)establish time alignment.
- RA-SR also is used, regardless of the uplink-timing state, when no PUCCH resources for D-SR were assigned to the UE.
- The SR procedure conveys little detail about the UE resource requirement, therefore
- ➤ a **buffer status report** (BSR) with more detailed information about the amount of data waiting in the UE
- is attached to the first uplink transmission following the SR procedure.



Scheduling for Quality of Service (QoS) - 1

- A particular challenge for the schedulers is to provide the desired quality of service (QoS) on a shared channel.
- GSM and UMTS provide guaranteed bit rates by preallocating radio resources statically to dedicated channels.
- LTE does not provide dedicated channels but only **shared channels.**

A number of default QoS characteristics suitable for:

- VoIP (Voice over the IP protocol)
- Signaling traffic
- Internet access
- **—** ...

have been standardized for EPS (Evolved Packet Services).

Scheduling for Quality of Service (QoS) - 2

- The required QoS characteristic are assigned by the EPC (Evolved Packet Core).
- It is the responsibility of the scheduler in the eNB to assign radio resources to satisfy the required characteristics.
- Depending on the implementation, the scheduler can base its scheduling decision on
 - the QoS class and the queuing delay of the available data
 - the instantaneous channel conditions
 - fairness indicators.

Summary of LT09-I & II

- Describe and revise the basic facts related to three coexisting technologies: GSM/EDGE, UMTS/HSPA, LTE
- Understand LTE architecture: EPC and E-UTRAN principles and functionalities at link-layer
 - UE and eNB
 - Serving Gateway S-GW
 - Packet Delivery Network PDN-GW
 - MME Mobility Management Entity
 - Today's tutorial- discussion of assignment 2 progress
 - Next lecture: LT10 on 5G and mesh network